



# Introduction

The FMOD Ex sound system is a revolutionary new audio engine for game developers, multimedia developers, sound designers, musicians and audio engineers, based on the years of experience of Firelight Technologies'™ previous product FMOD.

It also aims high - to push the boundaries of audio implementation for games and the like while at the same time using minimal resources and being scalable.

This new engine is written from the ground up since FMOD 3 was released and involves years of experience and feedback from FMOD users to create the most feature filled and easy to use product possible, without the drawbacks of legacy implementation that FMOD 3 may have suffered from its years of continuous development.

Some of the most exciting new features, which are described in more detail later are:

- Suite of **built in DSP special effects** which do not rely on any platform or operating system. (for a 100% cross platform audio experience). Includes **high quality I3DL2 compatible reverb!**
- **Next-gen console support.** PS3, Xbox 360 and Wii are fully supported.
- **Sound designer focus and tool.** The new suite of tools and functionality means FMOD is usable by sound designers and musicians and not just programmers. Sound authors will have the ability to create complex audio models and tweak them in real-time over the network (or even internet) while the game/application is still running!
- **Full 3D sound support** including linear/nonlinear/custom rolloff models, multiple listener support, occlusion and obstruction, sound cones, and support for stereo or multichannel samples being played in 3d!.
- **Geometry occlusion engine.** You can supply FMOD with a polygon scene and it will automatically occlude and obstruct direct path and reverb signals for you!
- **Virtual voices** to allow a game to play thousands of sounds at once on limited hardware without worrying about handling the logic to switch sounds off and on themselves.
- Support for **over 20 file formats.**

- **Advanced streaming engine** supporting gapless stitching/sentencing of sounds, low cpu overhead, multiple stream support, over-ridable file callbacks and more.
- **Compressed sample playback.** ADPCM, MPEG and XMA are able to be stored in memory without decompressing or streaming them, as if they were normal static samples!
- **Sub mixing** and channel groups.
- **2D / 3D sound morphing.** Set up a user supplied 5.1 or 7.1 2D mix, and morph between it and directional 3D sound! Great for entering and leaving volumetric sound sources.
- New **advanced 'DSP network' based software engine** to rival the most complex software synthesizer packages, all performed in real-time while the game/application is running! **Matrix panning** allows sound channels to be mapped to any speaker in any combination.
- New **object oriented API** supporting C, C++, C#, Delphi and Visual Basic.
- **Plug-in support** for ultimate flexibility. FMOD and VST plugins are supported. Everything in FMOD Ex has been designed with future expansion in mind.
- **SIMD optimized** (ie SSE, VMX, VFPU, ALTIVEC) mixing and filter routines for low cpu overhead. It is faster to use FMOD's software mixer than go through the driver overhead of DirectSound!

# Platform support

FMOD Ex supports the following hardware platforms. No other audio system available supports this many platforms.

- Microsoft Windows series.
- Microsoft Windows series 64bit. (AMD64)
- Linux
- Linux 64bit. (AMD64)
- Macintosh. OS8 / 9 / X and OSX for x86.
- Sony PlayStation 2
- Microsoft Xbox
- Nintendo Gamecube
- Sony Playstation Portable.
- Microsoft Xbox 360.
- PlayStation 3.
- Nintendo Wii.

That's currently 12 platforms! No other game audio library can claim to match anywhere near that many platforms!

# Feature list

## Unified API.

Samples, streams, music and CD API's are gone. Everything is now a 'Sound'

All types of sounds, including mods, midi files, wavs, oggs, samples, streams, cd tracks and fsb files can be accessed seamlessly through the one API.

## Virtual Channels

Virtual channels allow thousands of channels to play on limited hardware/software. Voices are swapped in and out according to 3d distance and priority.

### **Plug-in System**

New file formats, output modes, and encoders can be added or downloaded by the user as DLLs. VST and Winamp DSP plug-in support for effects is included.

### **Digital CD Playback**

Digital CDDA playback allows dsp effects / spectrum analysis, ripping etc just as if it was a normal PCM file being played back.

### **C++ API**

In FMOD Ex, a new C++ API is available as well as a standard C API.

All new FMOD API features are accessible through simple class types, such as the system class, sound class, channel class, DSP class. C/C++ headers naming conventions closely mapped.

For example - FMOD::System::init() in the C++ header would become FMOD\_System\_Init() in the C header.

### **C# and Visual Basic API**

FMOD Ex has full support for managed C# and Visual Basic interfaces

### **Multiple simultaneous soundcard support**

FMOD 3 was limited by only supporting 1 sound card at a time, so if you wanted to output to multiple cards at once you would have to instance fmod.dll multiple times.

Multiple output at once support is simply done by initializing multiple 'System' objects.

### **Multi-speaker output support**

Now FMOD has a full multichannel mixer, even 2D sounds can be played in 5.1 (or 7.1!). Sounds can even swap their channel assignments around so left and right of a stereo sound are swapped around, mixed or all placed in the rear left speaker for example.

The way this is available is FMOD supports pan matrices. Any input sound channel can be redirected to any output speaker, and on top of this percentages/fractional levels are supported, so there are no absolute speaker assignments.

Via ASIO, FMOD Ex now also supports full multichannel output access to up to 16 output channels for high end sound devices.

### **Multi-speaker input support**

Multichannel wavs, oggs and FSB files are supported for 5.1 music for example.

### **Low latency recording support**

FMOD Ex now supports super low latency recording, processing and output through a new recording engine.

Via ASIO the recording->DSP->playback latency can be as low as 1-3ms! This is great for realtime processing and playback of recorded audio.

### **Enhanced Internet features**

- Internet audio streaming. Custom internet streaming code is included, which allows for seamless SHOUTcast,

Icecast and http streaming support.

- Download capability. A side effect of FMOD™s modular file system which supports network files, even static samples can be loaded off the internet.

In fact you can use FMOD™s API to write an arbitrary file downloader!

- Voice chat In a future version, sever/client voice chat will be supported for real-time over the internet voice conversations! Compression such as SPEEX etc will be supported for low bandwidth.

### **File format support**

FMOD currently supports a wide range of audio file formats.

- AIFF - (Audio Interchange File Format)
- ASF - (Advanced Streaming format, includes support for the audio tracks in video streams)
- ASX - (playlist format - contains links to other audio files. To access contents, the FMOD Ex tag API is used)
- DLS - (DownLoadable Sound format for midi playback. Can also be used as a stand alone container format in FMOD)
- FLAC - (Lossless compression codec)
- FSB - (FMOD sample bank format generated by FSBank and FMOD designer tool)
- IT - (Impulse tracker sequenced mod format. FMOD Ex also fully supports resonant filters in .IT files, and the per channel or per instrument echo effect send, that can be enabled in ModPlug Tracker. This is **cross platform** effect support and does not require DirectX like other libraries do.)
- M3U - (playlist format - contains links to other audio files. To access contents, the FMOD Ex tag API is used)
- MID - MIDI using operating system or custom DLS patches.
- MOD - (Protracker / Fasttracker and others sequenced mod format)
- MP2 - (MPEG I/II Layer 2)
- MP3 - (MPEG I/II Layer 3, including VBR support)
- OGG - (Ogg Vorbis format)
- PLS - (playlist format - contains links to other audio files. To access contents, the FMOD Ex tag API is used)
- RAW - (Raw file format support. The user can specify the number of channels, bitdepth, format etc)
- S3M - (ScreamTracker 3 sequenced mod format)
- VAG - (PS2 / PSP format, playable on all platforms!)
- WAV - (Microsoft Wave files, including compressed wavs. PCM, MP3 and IMA ADPCM compressed wav files are supported across all platforms in FMOD Ex, and other compression formats are supported via windows codecs on that platform).
- WAX - (playlist format - contains links to other audio files. To access contents, the FMOD Ex tag API is used)
- WMA - (Windows Media Audio format)
- XM - (FastTracker 2 sequenced format)
- XMA - (Xbox 360 only)

File format plugins are also supported so the number of formats supported is limitless!

Note AAC is not included in FMOD Ex because the only reference source for this is GPL and FMOD Ex does not contain GPL protected code. To support this a user may add their own plugin to support it externally.

### **Wav Writer output**

All output can be written to a wav file, and with encoder plug-ins, it can even be encoded in real-time to MP3 or other file formats!

### **Sample accurate seeking**

Most systems seek to a compression block boundary such as mp3 which decodes in blocks of 1152 samples at a

time. FMOD Ex supports sample accurate seeking and decoding. For example you could seek to sample offset 1,000,000 exactly, and extract 1 sample of audio.

This accuracy is good for DJ type programs that need to sync streams properly.

### **Enhanced streaming engine**

A new low latency stream decoder that spreads the decode burden over time instead of doing it in chunks (cpu spikes!) is included. This means smoother frame-rates in game.

### **Enhanced sample format support**

24bit, 32bit integer and 32bit IEEE float sample support is included.

Alongside standard mono/stereo sample support, now multi-channel sample support is included!

Wav, ogg and user created sounds are examples of sound formats that support multi-channel sound.

### **Advanced mixing engine**

- **Enhanced output channel support**

Most systems only allow mixing to mono or stereo output. FMOD Ex allows mixing to any number of output channels, for example 6 channel output (with panning) to allow for 5.1 or Dolby digital output in real-time for 3d sound!

Stereo and 5.1 are optimized as a special case fast-path for extra speed.

- **Full DSP data flow network based mixing engine.**

New mixing routines with separate resample/mix/effects stages.

This is a node based multiple input/output DSP engine which is extremely flexible and allows submixing, splitting and advanced speaker location and selection.

- **High quality mixing**

All mixing is floating point with full 32bit interpolation.

Resampling modes supported are

- o No interpolation
- o Linear interpolation
- o Cubic interpolation
- o 5 point spline interpolation!

All resampling is done with true 32bit precision using a 32bit fractional, it is not downsampled or compromised in any way.

- **Matrix Panning**

Sounds can have their input channels mapped to any output channel through a simple 2D matrix. For example the left and right parts of a stereo sound can be positioned anywhere in a 5.1 speaker array, in any combination, in one speaker, or all speakers. It is totally flexible.

- **Volume ramping**

Linear volume ramps between pan/volume changes are included as standard. This removes clicks in sound that changes pan or volume frequently.

### **3D Sound enhancements**

- **Rolloff models.**

Logarithmic, linear, or custom rolloff models supported (per voice).

- **Geometry API.**

A revolutionary step up in audio realism is supported with FMOD Ex's custom geometry engine. This allows polygon scenes to be added to FMOD so that it can automatically calculate obstruction/occlusion as the user moves around the world.

- **Multiple listener support.**

Multiple 3d listeners for split screen support are supported.

- **Sound cone support.**

Sound cones are supported to give sounds direction.

- **Stereo / multichannel sound support.**

Stereo samples or even multichannel samples can be positioned in 3D, with their component channels (ie left/right parts of a stereo sound) positioned in 3D space, configurable by the user.

- **3D / 2D morphing.**

Now sounds can morph between being totally 3d directional point sources, and discrete 2D sources with speaker levels set by the user! This is great for entering and leaving a volumetric sound source. As an example, a stereo 3D sound can morph between being a directional point source, to a stereo 2D sound that envelopes you, then back again.

### **User delay on sound playback**

A new 'setDelay' function is available so a sound can be specified to start after a certain period of time (samples or ms) - can be called between init and start on a channel

### **MIDI Support**

FMOD Ex includes its own software midi playback, so that midi playback works cross platform.

Patch sets / DLS banks have to currently be provided with the song, or FMOD Ex will take advantage of any found in the operating system.

### **Stitching / sentencing**

Seamless stitching, for sounds allows one sound to end then another starts immediately afterwards without gaps. This is great for commentary or interactive music.

### **Built in software based special effects.**

FMOD Ex hosts a whole suite of special effects surpassing any system available considering it will work on every platform FMOD supports.

Here are some of the effects that will be supported as default. More can be added through plugins.

- Oscillators - sine, square, saw up, saw down, triangle and noise wave oscillators.
- 2 Low-pass with resonance filters.
- High-pass with resonance.
- 2 Echo filters.
- Flange.
- Distortion.
- Normalizer.
- Parametric EQ.
- Realtime pitch shifter (changes pitch not playback speed)
- Chorus.
- Freeverb simple reverb.
- SFX high quality I3DL2 compatible reverb.

### **Channel groups, and submixing.**

Multiple channel groups can be created and channels assigned to these groups.

From there a variety of commands can be issued on a group such as volume, mute, frequency, pause and more.

Master volume can be controlled through the use of a channel group, and multiple channel groups can be used for multiple master volume assignments, which is very useful for things like relative volume of GUI sounds vs in game sounds for example, or music vs special effects volume.

This allows greater flexibility in controlling audio levels.

Submixing allows effects to be placed on groups of channels, without affecting other channels. This is an advanced feature which is really useful for saving CPU usage or keeping some sounds dry while others are affected by DSP effects for example.

### **Enhanced callback support**

- 'latency adjusted' or 'real-time' flag for callbacks. This means you can get a callback at mix time, or audible time (the 2 are different, by the length of time determined by the mixer's buffer size)
- sample accurate user timer callbacks (ms or sample based) for global or per channel

### **Memory and filesystem overrides**

FMOD Ex of course allows the user to override FMOD's file and memory system through callbacks.

### **FMOD Designer tool and API**

- Sound Designer Tool

This easy to use and flexible sound designer tool allows simple or complex multi-layer/effect/envelope based sound events to be modeled and created by the sound designer. The capabilities would include such things as layering, effects, random behaviour, and stitching of sounds.

The aim is for a sound designer to totally design the in game audio from an external tool, and simply supply the programmer with assets and an event list to implement. If the audio behaviour needs to be changed within the game, it should be up to the sound designer not the programmer to do this.

The layering screen allows for complex audio models (such as a car engine with multiple cross fading channels, sounds and effects) to be totally controlled by the author, then all the programmer has to do is call the previously defined set of simple commands, such as SendEvent and UpdateParameter. In the car model cast, the "parameter"™ in UpdateParameter might just be "revs"™ or "torque"™ or some other English type value, rather than a value defined by a programmer.

- FMOD Event API This is an API for programmers to interface to the data produced by the FMOD Designer tool. This API consists of very simple commands such as:

- o Init
- o Close
- o Load
- o GetEvent / Start
- o UpdateParameter

All event behaviour is specified by the FMOD Designer tool, not the programmer, to make it totally data driven.

- Network tweaking features

As part of the sound designer tool, the user can tweak the audio parameters in a game over the network while the game is running! A sound designer now gets even more control over the outcome of the audio mix by being able to alter sound parameters such as volume / frequency / randomization etc while the game is running. This will save hours of time instead of the usual routine of testing, quitting, tweaking, recompiling, running. Even with that old method it can lead to mistakes which take several attempts to perfect. Using the network tweaking tool the sound designer can get it right first time.

# TERMINOLOGY / BASIC CONCEPTS.

## Introduction

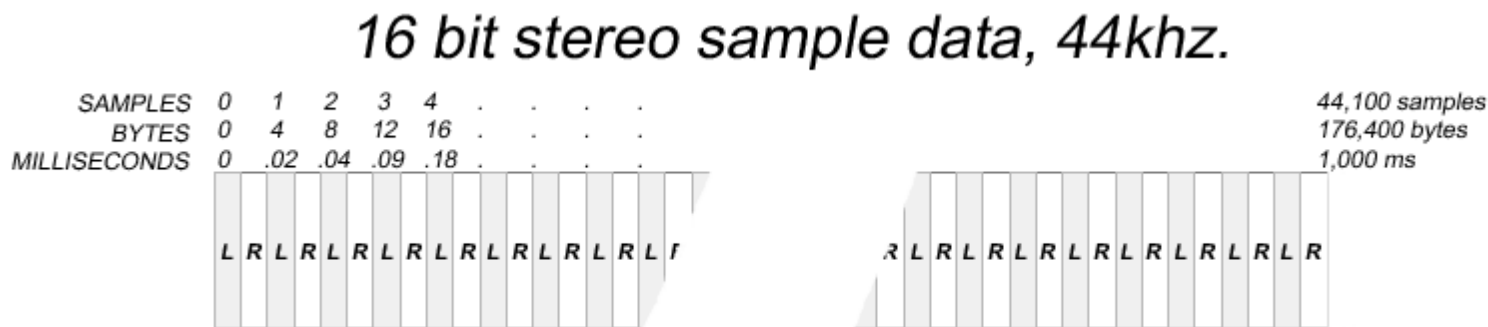
Throughout FMOD documentation certain terms and concepts will be used. This section will explain some of these to alleviate confusion.

It is recommended when you see an API function highlighted as a link, that you check the API reference for more detail.

## Samples vs bytes vs milliseconds

Within FMOD functions you will see references to PCM samples, bytes and milliseconds.

To understand what the difference is a diagram has been provided to show how raw PCM sample data is stored in FMOD buffers.



In this diagram you will see that a stereo sound has its left/right data interleaved one after the other.

- A left/right pair (a sound with 2 **channels**) is called a **sample**.
- Because this is made up of 16bit data, **1 sample = 4 bytes**.
- If the sample rate, or playback rate is 44.1khz, or 44100 samples per second, then **1 sample is 1/44100th of a second, or 1/44th of a millisecond**. Therefore 44100 samples = 1 second or 1000ms worth of data.

To convert between the different terminologies, the following formulas can be used.

- **ms** = samples \* 1000 / samplerate.
- **samples** = ms \* samplerate / 1000.
- **samplerate** = samples \* 1000 / ms.
- **bytes** = samples \* bits \* channels / 8.
- **samples** = bytes \* 8 / bits / channels.

Some functions like [Sound::getLength](#) provide the length in milliseconds, bytes and samples to avoid needing to do these calculations.

## Sounds. Samples vs compressed samples vs streams.

When a sound is loaded, it is either decompressed as a static sample into memory as PCM (samples), loaded into



memory in its native format and decompressed at runtime (compressed samples), or streamed and decoded in realtime (in chunks) from an external media such as a harddisk or CD (streams).

- **"Samples"** are good for small sounds that need to be played more than once at a time, for example sound effects. These generally use little or no CPU to play back and can be hardware accelerated. See [FMOD\\_CREATEAMPLE](#).
- **"Streams"** are good for large sounds that are too large to fit into memory and need to be streamed from disk into a small ringbuffer that FMOD manages. These take a small amount of CPU and disk bandwidth based on the file format. For example mp3 takes more cpu power to decode in real-time than a PCM decompressed wav file does. A streaming sound can only be played once, not multiple times due to it only having 1 file handle per stream and 1 ringbuffer to decode into. See [FMOD\\_CREATESTREAM](#).
- **"Compressed samples"** are a new advanced option that allows the user to load a certain compressed file format (such as IMA ADPCM, MP2, MP3 and XMA formats currently), and leave them compressed in memory without decompressing them. They are software mixed on the CPU and don't have the 'once only' limitation of streams. They take more cpu than a standard PCM sample, but actually less than a stream due to not doing any disk access and much smaller memory buffers. See [FMOD\\_CREATECOMPRESSEDAMPLE](#).

You may notice "Sample" and "Stream" terminology is used here but there is no class name with this terminology in them. That is because all FMOD APIs are now consolidated into one **"Sound"** type.

**By default [System::createSound](#) will want to decode the whole sound fully into memory** (ie, as a decompressed sample).

To have it stream in realtime and save memory, use the [FMOD\\_CREATESTREAM](#) flag when creating a sound, or use the helper function [System::createStream](#) which is essentially the same as [System::createSound](#) but just has the [FMOD\\_CREATESTREAM](#) flag added in automatically for you.

To make a compressed sample use [System::createSound](#) with [FMOD\\_CREATECOMPRESSEDAMPLE](#).

## Hardware vs Software

FMOD Ex has its support for either hardware accelerated sound playback, via DirectSound or console hardware API's, but FMOD also has its own fallback software mixing mechanism.

With hardware and software based sounds comes certain features and trade-offs when they are used.

Hardware sounds (created with [FMOD\\_HARDWARE](#) usually have lower CPU impact, have lower latency, and can get access to hardware reverb like EAX4 for example.

Hardware sounds are also limited in some ways, for example due to DirectSound limitations on Windows for example, arbitrary loop points are not supported with static samples (it is either loop the whole sound, or don't loop the sample), and non reverb effects cannot be played on them (ie chorus, distortion, lowpass etc).

Software sounds (created with [FMOD\\_SOFTWARE](#) sometimes have higher CPU impact, but can do much more, for example complex looping, realtime analysis, effects and sample accurate synchronization.

### Hardware vs Software.

#### Hardware Pros.

- Usually lower latency. (Although on consoles or ASIO output in windows, using FMOD\_SOFTWARE can have extremely low latency as low as 2-5ms)
- Less CPU time. (Although on Windows software is a lot faster due to bad hardware sound card driver design, and inefficiencies in the DirectSound API).
- On Windows, access to EAX2, EAX3, EAX4, I3DL2 reverb per voice. (FMOD Ex has its own high quality I3DL2 reverb solution in software, but may not be as flexible or have the quality of EAX4 for example.).
- Free hardware obstruction / occlusion (this is usually equivalent to a lowpass filter or reverb attenuation which can also be performed in software at some expense to the CPU), but only on EAX compatible sound cards on Windows. FMOD\_SOFTWARE is cross platform.
- On PS2, PSP, XBox, GameCube, Wii, hardware voices can play back ADPCM compressed sound data with

no cpu hit.

- On a limited number of soundcards, hardware 3d sounds will be realtime encoded into an AC3 Dolby Digital stream via a digital / optical output on the card so an amplifier can play it in 3D surround sound. FMOD software mixing now supports 5.1 and 7.1 mixing at slightly higher CPU expense, and will work via analog outputs such as soundcards with 3 stereo jacks to run to a 5.1 speaker setup.

### Hardware Cons.

- No point to point looping on win32. Xbox and GameCube allow point to point looping and PS2 only allows loopstart, so therefore cross platform compatibility cannot be assured.
- No access to hardware effects per voice. Most PC sound cards and consoles do not support hardware accelerated effects such as lowpass, distortion, flange, chorus etc.
- No loop count control. A sound can only be looped infinitely or not at all.
- Inconsistent feature support, for example a PS2 does not support EAX reverb, and 3d sound implementations always sound different.
- Sometimes a lot slower than FMOD software mixing on Windows. Virtual voices that make a lot of state changes when swapping in and out can be very expensive in hardware (noticeable framerate drops), but for free in software.

### Software Pros.

- Consistent sound on every platform, there is no variation in playback.
- Sample accurate synchronization callbacks and events.
- Compressed sample playback support without using streams.
- Cross platform reverb.
- Complex looping and loop counts.
- Reverse sample playback.
- Spectrum analysis.
- Filters per channel or for the global mix, to perform effects such as lowpass, distortion, flange, chorus etc.
- Complex DSP network construction for realtime sound synthesis.
- Access to final mix buffer to allow analyzing, drawing to screen, or saving to file.

### Software Cons.

- Latency on some sound devices (such as win32 waveout output) can be high.
- Memory usage is higher due to allocation of mix units and mix buffers, or simply the fact of having to store sounds in main ram rather than sound ram. (becoming less relevant these days).

## Channels and sounds.

When you have loaded your sounds, you will want to play them. When you play them you will use `System::playSound`, which will return you a pointer to a Channel / FMOD\_CHANNEL handle.

The index that [System::playSound](#) requires is generally recommended to always be [FMOD\\_CHANNEL\\_FREE](#). This will mean FMOD will choose a non playing channel for you to play on.

## 2D vs 3D.

A 3D sound **source** is a channel that has a position and a velocity. When a 3D channel is playing, its volume, speaker placement and pitch will be affected automatically based on the relation to the **listener**.

A **listener** is the player, or the game camera. It has a position, velocity like a sound **source**, but it also has an *orientation*.

The **listener** and the **source** distance from each other determine the *volume*.

The **listener** and the **source** relative velocity determines the *pitch* (doppler effect).

The orientation of the **listener** to the **source** determines the *pan* or *speaker placement*.

A 2D sound is simply different in that it is not affected by the 3D sound **listener**, and does not have doppler or attenuation or speaker placement affected by it.

A 2D sound can call [Channel::setSpeakerMix](#), [Channel::setSpeakerLevels](#) or [Channel::setPan](#), whereas a 3D sound cannot.

A 3D sound can call any function with the word **3D** in the function name, whereas a 2D sound cannot.

For a more detailed description of 3D sound, read the tutorial in the documentation on 3D sound.

# GETTING STARTED.

## Introduction

The FMOD Ex API has been designed to be intuitive and flexible. In this tutorial an introduction to using the engine as well as the key issues involved in using it effectively will be explained.

## Set up. What to include and what to link.

See "**Platform specific issues**" in this documentation to see what files to link into your project to make FMOD Ex function for each platform.

In C/C++, include "**fmod.h**" if you want to use the C interface only. Include "**fmod.hpp**" if you want to use the C++ interface.

Note that the constants, callbacks, defines and enums are stored within fmod.h, so fmod.hpp includes fmod.h. If you are using C++ you will be interchanging between both.

For Delphi, C# and Visual Basic, you will see equivalent headers to use in your application.

## Initialization.

The simplest way to initialize fmod is to simply call [System::init](#). That's it. FMOD will set up the soundcard and other factors using default parameters.

When looking at the documentation for [System::init](#), remember that the **maxchannels** parameter is the number of simultaneous voices you would like to be played in your game at once. This is nothing to do with how many hardware voices the soundcard may have, or how many software mixed voices there may be available.

These voices are [virtual voices](#). This means you can play as many sounds as you want at once and not worry about the issue of hardware or software resources available.

You can safely play EVERY sound in your game simultaneously without fear of [System::playSound](#) running out of voices or stealing other playing voices, and for this reason, it is acceptable to set maxchannels to a high number. 1, 100, 200, 1000. It is up to you and your type of title.

Note 1000 voices playing at once does not negatively impact performance because the majority of those will not be audible (non audible voices are 'virtualized'). There is only a small cost in sorting and swapping those voices as the FMOD Ex virtual voice manager controls which voices are heard and which aren't.

Let's have a look at an example of initializing FMOD Ex.

```
FMOD_RESULT result;
FMOD::System *system;

result = FMOD::System_Create(&system); // Create the main system object.
if (result != FMOD_OK)
{
    printf("FMOD error: (%d) %s\n", result, FMOD_ErrorString(result));
    exit(-1);
}

result = system->Init(100, FMOD_INIT_NORMAL, 0); // Initialize FMOD.
if (result != FMOD_OK)
{
    printf("FMOD error: (%d) %s\n", result, FMOD_ErrorString(result));
    exit(-1);
}
```

}

Here we have the most basic setup of the FMOD engine. It will use 100 virtual voices.

Note that mod/s3m/xm/it/midi formats use 1 voice when playing. Do not extend the voice count here thinking it will give more voices to these file formats when playing, because they won't. These formats have their own internal pool voices that they use.

## Configuration options

The output hardware, FMOD's resource usage, and other types of configuration options can be set if you desire behaviour differing from the default.

These are generally called before [System::init](#).

The main ones are.

- [System::setOutput](#) - To choose an alternative output method. For example you can choose between DirectSound, WinMM, ASIO, no-sound, wave-writer or a number of other output options in windows. Each platform will have their own output choices. Don't call this unless you need to. You don't need to call it especially if all you are doing is setting the default. That would be pointless.
- [System::setDriver](#) - To choose an alternative sound card driver for a particular output mode. This is useful if you have multiple sound cards and want to choose one beside the default. Again, don't bother calling this if all you are doing is setting it to the default. You should enumerate devices with [System::getNumDrivers](#) and [System::getDriverInfo](#) if you want to give the user the choice.
- [System::setHardwareChannels](#) - Call this if you want to limit the number of audible hardware voices, or request that a minimum number of hardware voices be available before reverting to 100% software mixed voice support. The 'minimum' option is to guarantee a certain number of voices are audible at once.
- [System::setSoftwareChannels](#) - Call this if you want to set a different number of audible software mixed voices used by FMOD Channels. This will be purely for polyphony reasons or CPU / memory resource usage reasons. Do not adjust this thinking it will give more voices to mod/s3m/xm/it/midi formats. They do not use this channel pool and have their own internally.
- [System::setSoftwareFormat](#) - Call this to change settings in the FMOD software mixer. This includes sample rate, output format (ie integer vs float), output channel count (ie for multi-output channel asio devices for example), memory usage and mixing quality.
- [System::setDSPBufferSize](#) - Call this only if there are issues with stuttering on slow machines or bad soundcard drivers. This will affect software mixing latency, and can have adverse effects if misused. Some titles may want to let the user select between 'low latency' and 'compatible' modes, so they can trade off latency to audible stability by adjusting the buffersize.
- [System::setSpeakerMode](#) - Call this to set the output speaker mode. This only affects the FMOD software mixing engine. The default is stereo (5.1 on xbox and xbox360 and 7.1 on ps3), and can be changed if desired. Note speaker modes with higher channel counts leads to higher memory usage.

Here is an example of initializing FMOD with some configuration options. **Remember these options are just that. Optional! Do not call these if you don't need to and don't just cut and paste this code without knowing what it does!** For example you can't just go setting the speaker mode to 5.1 if the user doesn't have a 5.1 speaker system!

```
FMOD_RESULT result;
FMOD::System *system;

result = FMOD::System_Create(&system); // Create the main system object.
CHECK(result);

result = system->setSpeakerMode(FMOD_SPEAKERMODE_5POINT1); // Set the output to 5.1.
CHECK(result);

result = system->setSoftwareChannels(100); // Allow 100 software mixed voices to be
audible at once.
CHECK(result);

result = system->setHardwareChannels(32, 64, 32, 64); // Equip the soundcard to handle
```

```
bas t3 2 2Da nd3 D h rwa e vices , a ndc h m pi t b usi g 64i fi t h s mo e th n ths .
E RE ECK ( esu lt;
```

```
esu lt= sys em- x m t(200, FMOD_IGN_TDRMAL, 0; // I n fa i l e MO D wi th 200 v r t u a l
vices .
E RE ECK ( esu lt;
```

## Loading and playing.

To play the sounds you must load them first!

To do this, use [System::createSound](#) or [System::createStream](#).

A sound by default will try to decompress the whole sound into memory (if [System::createSound](#) is used), that is why if the sound is large, it is better to stream it (by using [System::createStream](#)) which means it will decode at runtime, with a small fixed size memory buffer, and not use the memory a sample would.

For more on this see the [Terminology/Basic Concepts tutorial](#).

Here is an example of loading an mp3 file. By default [System::createSound](#) will decompress the whole MP3 into 16bit PCM. This could mean the amount of memory used is many times more than the size of the file.

```
MOD:Sou nd Sou nd
esu lt= sys em- x ea tSou nd( "../me dia/wa e m p ", FMOD_ E FAU LT, 0, ? // MO D E FAU LTuses
th d fa ul t. Th e a e th same as MO D DO PO FF | MO D 2D | MO D H RWA E .
E RE ECK ( esu lt;
```

Here is an example of opening an mp3 file to be streamed. [System::createStream](#) will open the file, and pre-buffer a small amount of data so that it will be able to play instantly when [System::playSound](#) is called.

```
MOD:Sou nd Sou nd
esu lt= sys em- x ea tS tream( "../me dia/wa e m p ", FMOD_ E FAU LT, 0, ? // MO D E FAU LT
uses th d fa ul t. Th e a e th same as MO D DO PO FF | MO D 2D | MO D H RWA E .
E RE ECK ( esu lt;
```

To specifically make a sound software mixed, you must use [FMOD\\_SOFTWARE](#). This is necessary if you want to use things such as DSP effects, spectrum analysis, getwavedata, point to point looping and other more advanced techniques.

```
MOD:Sou nd Sou nd
esu lt= sys em- x ea tSou nd( "../me dia/wa e m p ", FMOD_ S FWA E , 0, ? // Make th sou nd
so fwa e mi x d.
E RE ECK ( esu lt;
```

Here is an example of loading an mp3 file into memory as a sample, but not decompressing it when it loads, using the use [FMOD\\_CREATECOMPRESSED](#) flag. This will automatically make the sound software mixed if [FMOD\\_HARDWARE](#) or [FMOD\\_SOFTWARE](#) is not specified. *Hardware* sound playback cannot support this flag unless the format is ADPCM on Xbox, VAG on PS2/PSP and GCADPCM on Gamecube/Wii. Platforms like PS3 and Xbox 360 are all done on the cpu (usually a different core to the main cpu so it does not affect performance).

```
MOD:Sou nd Sou nd
esu lt= sys em- x ea tSou nd( "../me dia/wa e m p ", FMOD_ C ECOM PRESSE SAM PE , 0, ? //
FMOD_ C ECOM PRESSE SAM PE e l b th sam p e b a t e m p t p h y i g i t a s i t i s , w i t h u t
ecom pressi g i t i n b memo y f s t. Th s i s o n y su pp r e d f r IMA A DEM , M P2 , M B a n d
MA au d i o f ma t s .
E RE ECK ( esu lt;
```

**Warning!** This mode is to be used with care. It acts just like a PCM sample, but incurs a heavier CPU cost at runtime. FMOD decodes the sound from its compressed format as it plays it.

Now to play the sound or stream. This is as simple as calling [System:playSound](#).

```
FMOD: channel_t channel;  
result = sys_em->playSound(FMOD_CHANNEL_FREE, sound, false, ?  
ERR_CHECK(result);
```

This sound is now playing in the background! Your app will continue on from this point.

Things to note about playSound.

- **You do not need to store the channel handle** if you do not want to. That parameter can be 0 or NULL. This is useful if you don't care about updating that instance of the sound, and if it is a one shot sound (ie it does not loop). For example

```
FMOD: channel_t channel;  
result = sys_em->playSound(FMOD_CHANNEL_FREE, sound, false, 0;  
ERR_CHECK(result);
```

- **You can start the sound paused**, so you can update its attributes without the change being audible. That is what the 'paused' parameter is used for. For example, if you set it to true, set the volume to 0.5, then unpaused it, the sound would play at half volume. If you had set the paused flag to false and executed the same logic, you may hear the sound play at full volume for a fraction of a second. This can be undesirable.

```
FMOD: channel_t channel;  
result = sys_em->playSound(FMOD_CHANNEL_FREE, sound, true, ?  
ERR_CHECK(result);  
result = channel->setVolume(0.5f); // Set the volume while it is paused.  
ERR_CHECK(result);  
result = channel->setPaused(false); // This is where the sound really starts.  
ERR_CHECK(result);
```

- **A 'channel' is an instance of a sound.** You can play a sound many times at once, and each time you play a sound you will get a new channel handle. Not that this is only if it is not a stream. Streams can only be played once at a time, and if you attempt to play it multiple times, it will simply restart the existing stream and return the same handle that it was using before. This is because streams only have 1 stream buffer, and 1 file handle. To play a stream twice at once, open and play it twice.
- **Always use [FMOD\\_CHANNEL\\_FREE](#).** This lets FMOD pick the channels for you, meaning that it uses FMOD's channel manager to pick a non playing channel. [FMOD\\_CHANNEL\\_REUSE](#) can be used if the desired effect is to pass in an existing channel handle and use that for the playsound. It can be used to stop a sound spawning a new instance every time [System:playSound](#) is called, and only play once at a time.
- **You do not have to 'free' or 'release' a channel handle.** Channels come from a pool which you created by specifying a channel count in [System:init](#). Channel handles get re-used if old sounds have stopped on them. If all channels are playing, then one of the existing channels will get stolen based on the lowest priority sound. Make sure this doesn't happen by simply increasing the channel count in [System:init](#).
- **A channel becomes invalid once it is finished playing.** This means you can't update it, and doing so would be pointless anyway because it isn't going to start again. Referencing a stopped channel will most likely result in an [FMOD\\_ERR\\_INVALID\\_HANDLE](#).

## Update. (This is important!)

It is important that [System:update](#) be called once per frame. Do not call this more than once per frame, as this is not necessary and is just inefficient.

This function updates the following aspects of FMOD Ex.

- **Platform specific routines** such as the once a frame command packet send to the IOP on the PlayStation 2. Without the update no sound would be audible on this platform.
- **Virtual voice emulation.** Without update being called, virtual voices would pause.
- **3D voice calculation.** If update is not called, sounds will not audibly move in 3D even though the channel or listener has been had its 3D attributes set.
- **Geometry engine.** The FMOD polygon/geometry engine is updated from this function. Without it, the occlusion/obstruction properties defined by the user will not be audible.

- **Non realtime output.** [FMOD\\_OUTPUTTYPE\\_NOSOUND\\_NRT](#) and [FMOD\\_OUTPUTTYPE\\_WAVWRITER\\_NRT](#) need this function to be called to update to the output. (ie write to the file in [FMOD\\_OUTPUTTYPE\\_WAVWRITER\\_NRT](#)).
- **Streaming engine**, if [FMOD\\_INIT\\_STREAM\\_FROM\\_UPDATE](#) is specified. If the user has decided to drive the streaming engine themselves from the main thread, then update must be called regularly or the streamer will stutter and cause buffer underrun.

## Shutdown.

Call [System::release](#) to close the output device and free all memory associated with that object.

Channels are stopped, but sounds are *not* released. You will have to free them first. You do not have to stop channels yourself.

You can of course do it if you want, it is just redundant, but releasing sounds is good programming practice anyway.

You do not have to call [System::close](#) if you are releasing the system object. [System::release](#) internally calls [System::close](#) anyway.

## Resource usage configuration.

In application development, some developers will want to have all disk or memory access going through their own functions rather than using the default system.

In FMOD Ex, you can configure the FMOD file system to use your own file routines with [System::setFileSystem](#).

To make FMOD use your memory system, or to confine FMOD to 1 block of memory that it will not allocate outside of, use [Memory\\_Initialize](#).

**Note!** On Xbox and XBox 360 it is actually required for the user to provide FMOD with a block of memory. On Xbox 360 this memory must be allocated with **XPhysicalAlloc**. See "**Platform specific issues**" for more.



# TRANSITIONING BETWEEN FMOD 3 AND FMOD EX. API DIFFERENCES

## Introduction

This section will describe some of the differences between FMOD 3 and FMOD Ex, if you are used to the old API and have difficulty understanding the difference between the 2 APIs.

It will answer some of the more common questions usually beginning with "What happened to.."

## FMOD 3 had streams, sample and music APIs, now what?

All combined into the one class **Sound**. This leads to a much leaner and streamlined API. To create a stream just use [System::createStream](#) or FMOD\_CREATESTREAM flag with [System::createSound](#).

Music files loaded with the old music api would just be opened as a stream. As you could now load these types of sounds as a static sample (yes you can decode a whole mod into memory as PCM) it would possibly take hundreds of megabytes of ram, so even if you specify [System::createSound](#) to load a mod/s3m/xm/it file, it will still open it as a stream. To force it to a sample (not used as often) simply use FMOD\_CREATESAMPLE flag in

[System::createSound](#).

The old 'music' formats (mod/s3m/xm/it/midi) now being streams means you can also do cool things like place effects on music formats ([Channel::addDSP](#)), or treat them like a normal channel with [Channel::setVolume](#) / [Channel::setFrequency](#) etc and therefore can even 3d position them!.

## Channels are now objects instead of just integer handles

FMOD Ex now takes a more object oriented approach than FMOD 3. Channel objects are still reference counted though! So if the channel handle you have is stolen, FMOD Ex will still know not to update the newly playing channel with commands issued from the old channel handle.

Channel stealing should be less prevalent now thanks to virtual channels. You can now allocate a pool of many hundreds or even thousands of channels which will never run out, and they all succeed when you try to play them all at once. This is thanks to FMOD Ex's new **virtual voice system**.

## Volume and pan

FMOD 3 used volume 0-255 (silent to full volume) and pan 0-255 (left to right), but now FMOD Ex takes a floating point number for each.

FMOD Ex now uses **0.0 to 1.0** for volume (silent to full), and **-1.0 to +1.0** for pan (left to right, 0.0 = center).

## Frequency

FMOD 3 used integer frequencies. FMOD Ex now uses floating point frequencies. Now you can get far greater accuracy for sound playback (ie you can now set 44100.5 instead of having to choose between 44100 and 44101) which is important when trying to do exact playback synchronization between 2 streams of different bpm for

example.

## **FSOUND\_GetError is gone**

FMOD 3 used a global error code for determining what an error was. This was a pretty bad design choice, as internal and multithreaded FMOD calls could contaminate the global error code.

FMOD Ex now uses a much cleaner error return code for every single function. This is not affected by the previously mentioned issues.

## **What happened to FSOUND\_SetSFXMasterVolume? or How do I perform master volume?**

FMOD 3 used this function to scale all non music oriented channel volumes.

FMOD Ex now uses 'ChannelGroups' which are far more powerful, and to scale all channels by a master volume, just use [System::getMasterChannelGroup](#) then [ChannelGroup::setVolume](#).

Using ChannelGroups you can now have multiple master volume groups, and other exciting features such as DSP submixing.

## **MOD/S3M/XM/IT channels used to take up channels in FMOD 3's main channel pool so I had to adjust FSOUND\_Init, do I have to do this with [System::init](#)?**

No. MOD/S3M/XM/IT (and now MIDI) have their own channel pools that do not affect the number in [System::init](#). Just select a number of channels that YOU are going to use, don't worry about what FMOD Ex is doing internally. To you playing a MOD/S3M/XM/IT/MID uses 1 channel.

## **Where is FSOUND\_SetHWND?**

FMOD Ex is global focus, or windowless by default. If you really need to focus the audio on a particular window in FMOD\_OUTPUTTYPE\_DSOUND mode on win32/win64, pass the hwnd as the extradrivervdata parameter in [System::init](#)

## **Where is FSOUND\_GetCurrentLevels?**

Use [System::getWaveData](#) or [Channel::getWaveData](#). It is far more flexible.

## **FSOUND\_Update is now System::update.**

Call [System::update](#) once a frame in your game loop. This is necessary to update various aspects of FMOD Ex.

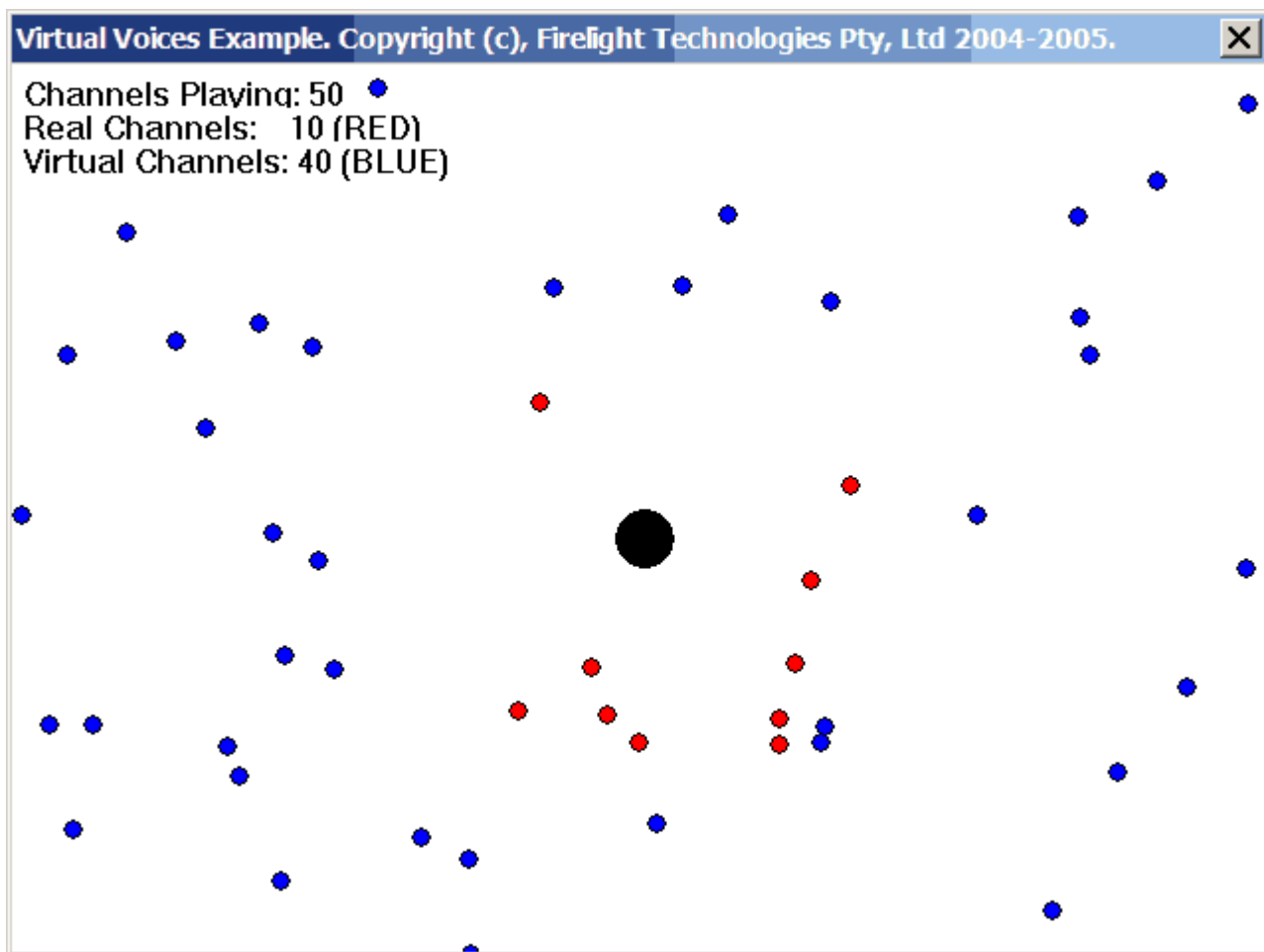
# CHANNEL MANAGEMENT AND VIRTUAL VOICES.

## Introduction

FMOD Ex now includes an efficient virtual voice management system. This tutorial will explain how it works and what is the advantage of using virtual voices.

## What are virtual voices?

What is a virtual voice and how is this different to a hardware or software voice?



Notice this screenshot of the FMOD Ex virtual voices example. It is playing 50 sounds at once, but only 10 are audible.

On limited sound hardware, which generally only has 32 to 64 voices, it can be challenging to manage your whole game's audio voice allocation when you want to have hundreds or even thousands of sounds playing at once in world (for example in a dungeon there might be 200 torches burning on walls in various places all playing a crackling burn noise).

FMOD Ex now allows the user to play as many sounds as they require, and will automatically allocate the limited number of hardware or software voices to the most important sounds to the listener.

This could mean in as in the above example, in a 3D world the 10 closest sounds are audible and the rest become

'virtual'.

Notice in the above screenshot red sounds are audible, and the blue sounds are inaudible and 'virtual'. According to the user though, there are actually 50 sounds playing at once.

'Virtual' voices are not allocated to a hardware or software voice, and are usually the least important sounds to the listener. These are 'emulated' voices.

They will update their play cursors and seem to be playing like a normal sound, but will not be audible.

As the user moves around the world, or a 'virtual' voice suddenly becomes more important than one that is actually audible, FMOD's virtual voice manager will swap the two voices, and the sound that was previously virtual will now become audible at its correct position in time.

A voice can be queried if it is virtual or not by using the [Channel::isVirtual](#) function. This is usually only for informational purposes.

## What if some sounds are more important than others?

First we will take the case of 2D sounds that are all playing at the same volume. How do you make sure one sound stays audible and the others possibly become virtual if too many sounds are playing?

The answer is to use the [Channel::setPriority](#) or [Sound::setDefault](#) function.

By making one sound have a higher priority than another, it will be given priority to be audible while its competitor will be swapped out and become virtual.

For example if there were 10 sounds playing and only 10 real voices, and an 11th voice wants to be played. If the new sound has a higher priority than the voices playing, it will be played as audible and one of the original 10 will become virtual, because the new sound is more important.

Important sounds should have higher priority and it is up to the user to decide if some sounds should be more important than others. An example of an important sound might be a 2D menu or GUI sound or beep that needs to be heard above all other sounds.

Volume of a sound is a secondary determining factor between sounds of equal priority. If a group of sounds have the same priority, the loudest sound will be the most important. In a 3d world this usually means the closest sounds will be more important and the further away sounds, or the quieter sounds will be less important and will possibly become virtual.

## What if I run out of virtual voices?

If you try to play more sounds than there are virtual voices, then FMOD Ex channel manager will try to find the least important sound and replace it with the new sound. This means the channel that has been replaced will stop and become invalid.

If a channel handle that has been kicked out by a new channel becomes invalid, any commands that are used on that channel handle will return [FMOD\\_ERR\\_INVALID\\_HANDLE](#).

## How do I set the number of real voices and virtual voices?

To set the number of virtual voices FMOD Ex will use, call [System::init](#) with the number of virtual voices specified in the maxchannels parameter.

For hardware voices, generally you don't set the number of these available on a sound device, such as on a console or sound card. Usually you are provided with a number of hardware channels to use. For example, PlayStation 2 always has 48 hardware voices.

On a sound card, this is variable depending on the manufacturer. You can find out the number of available hardware channels with [System::getHardwareChannels](#).

If you want to limit the number of hardware channels below its capacity, you can use [System::setHardwareChannels](#). This type of voice is used if the sound is created with [FMOD\\_HARDWARE](#) flag.

To set the number of software mixed channels available, use [System::setSoftwareChannels](#). You can set this to 0 if you don't want any software mixed voices.

This type of voice is used if the sound is created with [FMOD\\_SOFTWARE](#) flag.

## How many virtual voices should I set?

How many sounds are you trying to play at once without losing control of the channel handles?

This figure is up to you, but remember that more channels = more CPU and memory usage.

If you have 32 real hardware or software channels available to you and don't want to play more than this at once, then you might only need 32 virtual voices. This will mean a 1 to 1 relationship between real voices and virtual voices and sounds will never become emulated and be swapped out. Instead if you play more than the specified amount of channels, it will 'kick out' other lower priority channels.

If you have 32 real hardware or software channels available and you want to be able to safely play 100 at once, or 1000 at once, then set it to 100 or 1000 at once. Figures around the 1000 mark playing at once might start to show non negligible amounts of CPU and memory usage so be wary of this. Use [System::getCPUUsage](#) and [FMOD::Memory\\_GetStats](#) to determine this.

## Can I make silent sounds go virtual?

Yes. To do this enable the [FMOD\\_INIT\\_VOL0\\_BECOMES\\_VIRTUAL](#) flag in [System::init](#).

To configure this even further, you can change it from volume 0, to a higher volume, between 0 and 1. For example if you set the level for voices to go virtual at 0.1, everything below this audibility would go virtual. Warning if this is set too high, sounds may appear to 'cut out' before they are silent.

Use [System::setAdvancedSettings](#), and the 'vol0virtualvol' member of [FMOD\\_ADVANCEDSETTINGS](#).

## How do I tell if a Channel is virtual or not?

See [Channel::isVirtual](#)

## What if I don't like the sound of a voice going from virtual to real and playing half way through the sound, or near the end of the sound? (sounds like a bug!)

You can either use Sound or Channel priorities to stop it going virtual in the first place, or you have the option to have a voice start a from the beginning instead of half way through, by using the [FMOD\\_VIRTUAL\\_PLAYFROMSTART](#) flag with [System::createSound](#), [System::createStream](#), [Sound::setMode](#) or [Channel::setMode](#).

As described above, only the quietest, least important sounds should be swapping in and out, so you shouldn't notice sounds 'swapping in', but if you have a low number of real voices, and they are all loud, then this behaviour could become more noticable and may sound bad.

Another option is to simply call [Channel::isVirtual](#) and stop the sound, but don't do this until after a [System::update!](#)

After playsound, the virtual voice sorting needs to be done in [System::update](#) to process what is really virtual and what isn't.



# 3D SOUND

## Introduction.

This section will introduce you to using 3D sound with FMOD Ex. With it you can easily implement interactive 3D audio and have access to features such as 5.1 or 7.1 speaker output, and automatic attenuation, doppler and more advanced psychoacoustic 3D audio techniques.

## Loading sounds as '3D'.

When loading a sound or sound bank, the sound must be created with [System::createSound](#) or [System::createStream](#) using the [FMOD\\_3D](#) flag.  
ie.

```
result = system->createSound("../media/drumbo.wav", FMOD_3D, 0, ?  
if (result != FMOD_OK)  
{  
    HandleError(result);  
}
```

This will try and allocate a sound using hardware mixing by default. If there is no hardware mixing available, it will use software mixing as fallback.

To specifically load a sound in hardware or software simply add [FMOD\\_HARDWARE](#) or [FMOD\\_SOFTWARE](#)  
ie.

```
result = system->createSound("../media/drumbo.wav", (FMOD_MODE) (FMOD_HARDWARE |  
FMOD_3D), 0, ?  
if (result != FMOD_OK)  
{  
    HandleError(result);  
}
```

Note that once the sound is loaded, on Win32 and [FMOD\\_OUTPUTTYPE\\_DSOUND](#) output (the default on Win32), you can't change the mode from [FMOD\\_3D](#) to [FMOD\\_2D](#) and vice versa. This is a limitation of DirectSound.

Using [FMOD\\_SOFTWARE](#) instead of [FMOD\\_HARDWARE](#) alleviates this issue, and other platforms that support hardware (ie Xbox, PS2, Gamecube) allow switching between 2D and 3D.

It is generally best not to try and switch between 3D and 2D at all, if you want though, you can change the sound or channel's mode to [FMOD\\_3D\\_HEADRELATIVE](#) at runtime which places the sound always relative to the listener, effectively sounding 2D as it will always follow the listener as the listener moves around.

## Distance models and linear rolloff vs logarithmic.

### Logarithmic

This is the default FMOD 3D distance model. All sounds naturally attenuate (fade out) in the real world using a logarithmic attenuation. The flag to set to this mode is [FMOD\\_3D\\_LOGROLLOFF](#) but if you're loading a sound you don't need to set this because it is the default. It is more for the purpose of resetting the mode back to the original if you set it to [FMOD\\_3D\\_LINEARROLLOFF](#) at some later stage.

When FMOD uses this model, '**mindistance**' of a sound / channel, is the distance that the sound *starts* to attenuate from. This can simulate the sound being smaller or larger. By default, for every doubling of this mindistance, the sound volume will halve. This rolloff rate can be changed with [System::set3DSettings](#).

As an example of relative sound sizes, we can compare a bee and a jumbo jet. At only a meter or 2 away from a bee we will probably not hear it any more. In contrast, a jet will be heard from hundreds of meters away. In this case we might set the bee's mindistance to 0.1 meters. After a few meters it should fall silent. The jumbo jet's mindistance could be set to 50 meters. This could take many hundreds of meters of distance between listener and sound before it falls silent. In this case we now have a more realistic representation of the loudness of the sound, even though each wave file has a fully normalized 16bit waveform within. (ie if you played them in 2D they would both be the same volume).

The '**maxdistance**' does not affect the rate of rolloff, it simply means the distance where the sound *stops* attenuating. **Don't set the maxdistance** to a low number unless you want it to artificially stop attenuating. This is usually not wanted. Leave it at its default of 10000.0.

## Linear

This is an alternative distance model that FMOD has introduced. It is supported by adding the [FMOD\\_3D\\_LINEARROLLOFF](#) flag to [System::createSound](#) or [Sound::setMode](#) / [Channel::setMode](#). This is a more fake, but usually more game programmer friendly method of attenuation. It allows the '**mindistance**' and '**maxdistance**' settings to change the attenuation behaviour to fading linearly between the two distances. Effectively the mindistance is the same as the logarithmic method (ie the minimum distance before the sound starts to attenuate, otherwise it is full volume), but the maxdistance now becomes the point where the volume = 0 due to 3D distance. The attenuation inbetween those 2 points is linear.

## Some global 3D settings.

The 3 main configurable settings in FMOD Ex that affect all 3D sounds are:

- Doppler factor. This is just a way to exaggerate or minimize the doppler effect.
- Distance factor. This allows the user to set FMOD to use units that match their own (ie centimeters, meters, feet)
- Rolloff scale. Affects 3d sounds that use FMOD\_3D\_LOGROLLOFF. Controls how fast all sounds attenuate using this mode.

All 3 settings can be set with [System::set3DSettings](#). Generally the user will not want to set these.

## Velocity and keeping it frame rate independent.

Velocity is only required if you want doppler effects. Otherwise you can pass 0 or NULL to both [System::set3DListenerAttributes](#) and [Channel::set3DAttributes](#) for the velocity parameter, and no doppler effect will be heard.

This must be stressed again. It is important that the velocity passed to FMOD Ex is meters **per second** and not meters **per frame**. Notice the difference.

To get the correct velocity vector, use vectors from physics code etc, and don't just subtract last frames position from the current position. This is affected by framerate. The higher the framerate the smaller the position deltas, and therefore smaller doppler effects, which is incorrect.

If the only way you can get the velocity is to subtract this and last frame's position vectors, then remember to time adjust them from meters per frame back up to meters per second.

This is done simply by scaling the difference vector obtained by subtracting the 2 position vectors, by one over the frame time delta.

Here is an example.



```

float lx = (ps.x - hs.tps.x) * 1000 / timeDelta;
float lz = (ps.z - hs.tps.z) * 1000 / timeDelta;
float lz = (ps.z - hs.tps.z) * 1000 / timeDelta;

```

timeDelta is the time since the last frame in milliseconds. This can be obtained with functions such as `timeGetTime()`. So at 60fps, the timeDelta would be 16.67ms. if the source moved 0.1 meters in this time, the actual velocity in meters per second would be:

```

float v = 0.1 * 1000 / 16.67 = 6 meters per second.

```

Similarly, if we only have half the framerate of 30fps, then subtracting position deltas will give us twice the distance that it would at 60fps (so it would have moved 0.2 meters this time).

```

float v = 0.2 * 1000 / 33.33 = 6 meters per second.

```

## Orientation and left-handed vs right-handed coordinate systems.

Getting the correct orientation set up is essential if you want the source to move around you in 3d space. FMOD Uses a left handed coordinate system by default, (+X = right, +Y = up, +Z = forwards), which is the same as DirectSound3D and A3D.

If you use a different coordinate system, then you will need to flip certain axis or even swap them around inside the call to [System::set3DListenerAttributes](#) and [Channel::set3DAttributes](#).

Take the right handed coordinate system, where +X = right, +Y = up, +Z = backwards or towards you. To convert this to FMOD coordinate system simply negate all instances of the Z coordinate for listener and sound position and velocity, as well as listener up and forward vector Z components.

To make things easier for people using the right handed coordinate system, you can initialize FMOD Ex using [FMOD\\_INIT\\_3D\\_RIGHTHANDED](#) in [System::init](#) and not do any conversion. FMOD will automatically convert its internal 3D calculations to be right handed instead of left handed.

## A typical game loop.

3D sound and the FMOD channel management system need to be updated once per frame.

To do this use [System::update](#)

This would be a typical example of a game audio loop.

```

do
{
    UpdateGame(); // here the game is updated and the sources would be moved with
    camera's position.

    system->set3DListenerAttributes(0, ? // update 'ears'

    system->update(); // need to update 3d engine, once per frame.

} while (game running);

```

Most games usually take the position, velocity and orientation from the camera's vectors and matrix.

## Stereo and multichannel sounds can be 3D!

A stereo sound when played as 3d, will be split into 2 mono voices internally which are separately 3d positionable. Multi-channel sounds are also supported, so an 8 channel sound for example will allocate 8 mono voices internally in FMOD.

To rotate the left and right part of the stereo 3d sound in 3D space, use the [Channel::set3DSpread](#) function.

By default the subchannels position themselves in the same place, therefore sounding 'mono'.

## Split screen / multiple listeners.

In some games, there may be a split screen mode. When it comes to audio, this means that FMOD Ex has to know about having more than 1 listener on the screen at once.

This is easily handled via [System::set3DNumListeners](#) and [System::set3DListenerAttributes](#).

If you have 2 player split screen, then for each 'camera' or 'listener' simply call [System::set3DListenerAttributes](#) with 0 as the listener number of the first camera, and 1 for the listener number of the second camera.

[System::set3DNumListeners](#) would be set to 2.

That's all there is to it. You may notice an audible difference, because fmod does a few things to avoid confusion with the same sound being viewed from different viewpoints.

- 1. It turns off all doppler. This is because one listener might be going towards the sound, and another listener might be going away from the sound. To avoid confusion, the doppler is simply turned off.
- 2. All audio is mono. If to one listener the sound should be coming out of the left speaker, and to another listener it should be coming out of the right speaker, there will be a conflict, and more confusion, so all sounds are simply panned to the middle. This removes confusion.
- 3. Each sound is played only once as it would with a single player game, saving voice and cpu resources. This means the sound's effective audibility is determined by the closest listener to the sound. This makes sense as the sound should be the loudest to the nearest listener. Any listeners that are further away wouldn't have any impact on the volume at this point.

## Speaker modes / output.

To get 5.1 sound is easy. If the sound card supports it, then any sound using [FMOD\\_3D](#) and [FMOD\\_HARDWARE](#) will automatically position itself in a surround speaker system, and only the user has to be sure that the speaker settings in the operating system are correct so that the sound device can output the audio in 5.1 or 7.1.

**You do not need to call [System::setSpeakerMode](#)!.** This function is only used to configure FMOD Ex's software mixing engine. See the next paragraph on this.

For sounds created with [FMOD\\_SOFTWARE](#), by default sound is emulated through a simple stereo output. This involves panning and volume attenuation.

To enable FMOD software mixing to use 5.1 output, you can use [System::setSpeakerMode](#). But note! This function increases the CPU mixing burden slightly as it now has to software mix into a 6 or 8 channel buffer instead of a stereo buffer.

# FMOD\_NONBLOCKING flag and asynchronously loading data

## Introduction

[FMOD\\_NONBLOCKING](#) flag is used so that sounds can be loaded without affecting the framerate of the application.

Normally loading operations can take a large or significant amount of time, but with this feature, sounds can be loaded in the background without the application skipping a beat.

## Creating the sound.

Simply create the sound as you normally would but add the [FMOD\\_NONBLOCKING](#) flag.

```
FMOD: Sound Sound
result = sys_emsoundteam("../media/wave.mp3", FMOD_NONBLOCKING, 0, ? // Creates a
handle to a stream that commands the MODAsynchronous loader to open the stream in the
background.
```

```
ERRCHECK(result);
```

Now the sound will open in the background, and you will get a handle to the sound immediately. You cannot do anything with this sound handle except call [Sound::getOpenState](#). Any other attempts to use this sound handle will result in the function returning [FMOD\\_ERR\\_NOTREADY](#).

## Getting a callback when the sound loads.

When the sound loads or the stream opens, you can specify a callback using the nonblockcallback member of the [FMOD\\_CREATESOUNDEXINFO](#) structure to make the non-blocking open seek to the subsound of your choice. Firstly the callback definition.

```
FMOD_RESULT CALLBACK nonblockcallback(FMOD_SOUND Sound, FMOD_RESULT result)
{
    FMOD: Sound Sound = (FMOD: Sound)Sound;

    printf("Sound loaded (%d) %s\n", result, FMOD_ErrorString(result));

    return FMOD_OK;
}
```

And then the createSound call.

```
FMOD_RESULT result
FMOD: Sound Sound
FMOD_CREATESOUNDEXINFO exinfo;

memset(&exinfo, 0, sizeof(FMOD_CREATESOUNDEXINFO));
exinfo.nonblockcallback = nonblockcallback;

result = sys_emsoundteam("../media/wave.mp3", FMOD_NONBLOCKING, ?
ERRCHECK(result);
```

## Waiting for the sound to be ready and using it.

As mentioned, you will have to call [Sound::getOpenState](#) to wait for the sound to load in the background. You could

do this, or just continually try to call the function you want to call (ie [System::playSound](#)) until it succeeds. Here is an example of polling the sound until it is ready, then playing it.

```
FMOD_RESULT result;
FMOD::Sound *sound;

result = system->createStream("../media/war.mp3", FMOD_CREATESTREAM, 0, 0, // Creates a
blocking stream that commands the FMOD asynchronous loader to open the stream in the
background.

while (result != FMOD_OK);

do
{
    FMOD_DOESSTATE state;

    result = system->getState(&state);
    while (result != FMOD_OK);

    if (state == FMOD_DOESSTATE_READY){
        result = system->playSound(FMOD_CREATESTREAM, sound, false, 0);
        while (result != FMOD_OK);
    }

    GameComplete();
} while (1)

or

do
{
    if (change)
    {
        result = system->playSound(FMOD_CREATESTREAM, sound, false, 0);
        if (result != FMOD_OK)
        {
            while (result != FMOD_OK);
        }
    }
}

GameComplete();
} while (1)
```

The second loop will simply retry playsound until it succeeds.

## Creating the sound as a streamed FSB file.

An FSB file will have subsounds in it, so if you open it as a stream, you may not want FMOD seeking to the first subsound and wasting time. You can use the `initialsubsound` member of the [FMOD\\_CREATEINDEXINFO](#) structure to make the non-blocking open seek to the subsound of your choice.

```
FMOD_RESULT result;
FMOD::Sound *sound;
FMOD_CREATEINDEXINFO info;

memset(&info, 0, sizeof(FMOD_CREATEINDEXINFO));
info.initialsubsound = 1;

result = system->createStream("../media/sound.fsb", FMOD_CREATESTREAM, 0, 0, &info);
while (result != FMOD_OK);
```

Then get the subsound you wanted with [Sound::getSubSound](#).

## Getting a subsound.

[Sound::getSubSound](#) is a free function call normally, all it does is return a pointer to the subsound, whether it be a sample or a stream. It does not execute any special code besides this.

What it would cause if it was a blocking stream though, is [System::playSound](#) stalling several milliseconds or more while it seeks and reflushes the stream buffer. Time taken can depend on the file format and media.

If the parent sound was opened using [FMOD\\_NONBLOCKING](#), then it will set the **subsound** to be [FMOD\\_OPENSTATE\\_SEEKING](#) and it will become not ready again until the seek and stream buffer flush has completed.

When the stream is ready and [System::playSound](#) is called, then the playsound will not stall and will execute immediately because the stream has been flushed.

# MEMORY MANAGEMENT AND CONSERVATION TUTORIAL

## Introduction

This section will give some pointers on how to use and save memory in FMOD Ex by describing things that may not be so obvious upon first looking at the API.

## Using a fixed size memory pool.

To make FMOD stay inside a fixed size memory pool, and not do any external allocs, you can use the [FMOD::Memory\\_Initialize](#) function.

i.e.

```
result = FMOD::Memory_Initialize( malloc( 4*1024*1024 , 4*1024*1024, 0,0,0 ); //  
allocate bandwidth pass it to FMOD Ex to use .  
CHECK( result;
```

**Note** that this uses malloc. On Xbox 360 and Xbox you must use a different operating system alloc such as XPhysicalAlloc otherwise FMOD may not behave correctly. See "Platform specific issues" tutorials for more information on this.

Note that this function allows you to specify your own callbacks for alloc and free. In this case the memory pool pointer and length must be NULL. The 2 features are mutually exclusive.

## Lowering sound instance overhead.

The [FMOD\\_LOWMEM](#) flag is used for users wanting to shave some memory usage off of the sound class. This flag removes memory allocation for certain features like the 'name' field which isn't used often in games. When this happens, [Sound::getName](#) will return "(null)".

More memory will be stripped from the sound class in future versions of FMOD Ex when this flag is used. Currently the 'name' field is the biggest user of memory in the sound class so this has been removed first.

## Using compressed samples.

To trade CPU usage vs Memory, FMOD Ex has a feature to play ADPCM, XMA and MP2/MP3 data compressed, without needing to decompress it to PCM first. This can save a large amount of memory. On Xbox 360, using this for XMA files incurs next to no extra CPU usage, as the Xbox 360 XMA hardware decoder does the data decompression in realtime.

To enable this use the [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) flag. If this flag is used for formats other than the ones specified above, it will be ignored.

With the exception of XMA on Xbox 360 and ADPCM on Xbox, if [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) is used with an FMOD\_HARDWARE buffer it will generate an [FMOD\\_ERR\\_NEEDSSOFTWARE](#) error.

**Note!** If you use [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) there will be a 'one off' memory overhead to allocate the appropriate pool of codecs depending on the format being loaded. See the next section on how to control this

pool.

## Controlling memory usage with settings.

- [System::setSoftwareFormat](#) 'maxinputchannels' is default to 6 to allow up to 6 channel wav files to be played through FMOD's software engine. Setting this to a lower number will save memory across the board. If the highest channel count in a sound you are going to use is stereo, then set this to 2.
- For sounds created with [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#), [System::setAdvancedSettings](#) allows the user to reduce the number of simultaneous XMA/ADPCM or MPEG sounds played at once, to save memory. The defaults are specified in the documentation for this function. Lowering them will reduce memory. Note the pool of codecs for each codec type is only allocated when the first sound of that type is loaded. Reducing XMA to 0 when XMA is never used will not save any memory.
- For streams, setting [System::setStreamBufferSize](#) will control the memory usage for the stream buffer used by FMOD for each stream. Lowering the size in this function will reduce memory, but may also lead to stuttering streams. This is purely based on the type of media the FMOD streamer is reading from (ie CDROM is slower than harddisk), so it is to be experimented with based on this.
- Reducing the number of channels used will reduce memory. [System::init](#) and [System::setSoftwareChannels](#) give control over maximum number of virtual voices and software voices used. You will need to make sure you specify enough voices though to avoid channel stealing.

## Tracking FMOD memory usage.

Using [FMOD::Memory\\_GetStats](#) is a good way to track FMOD memory usage, and also find the highest amount of memory allocated at any time, so you can adjust the fix memory pool size for the next time.

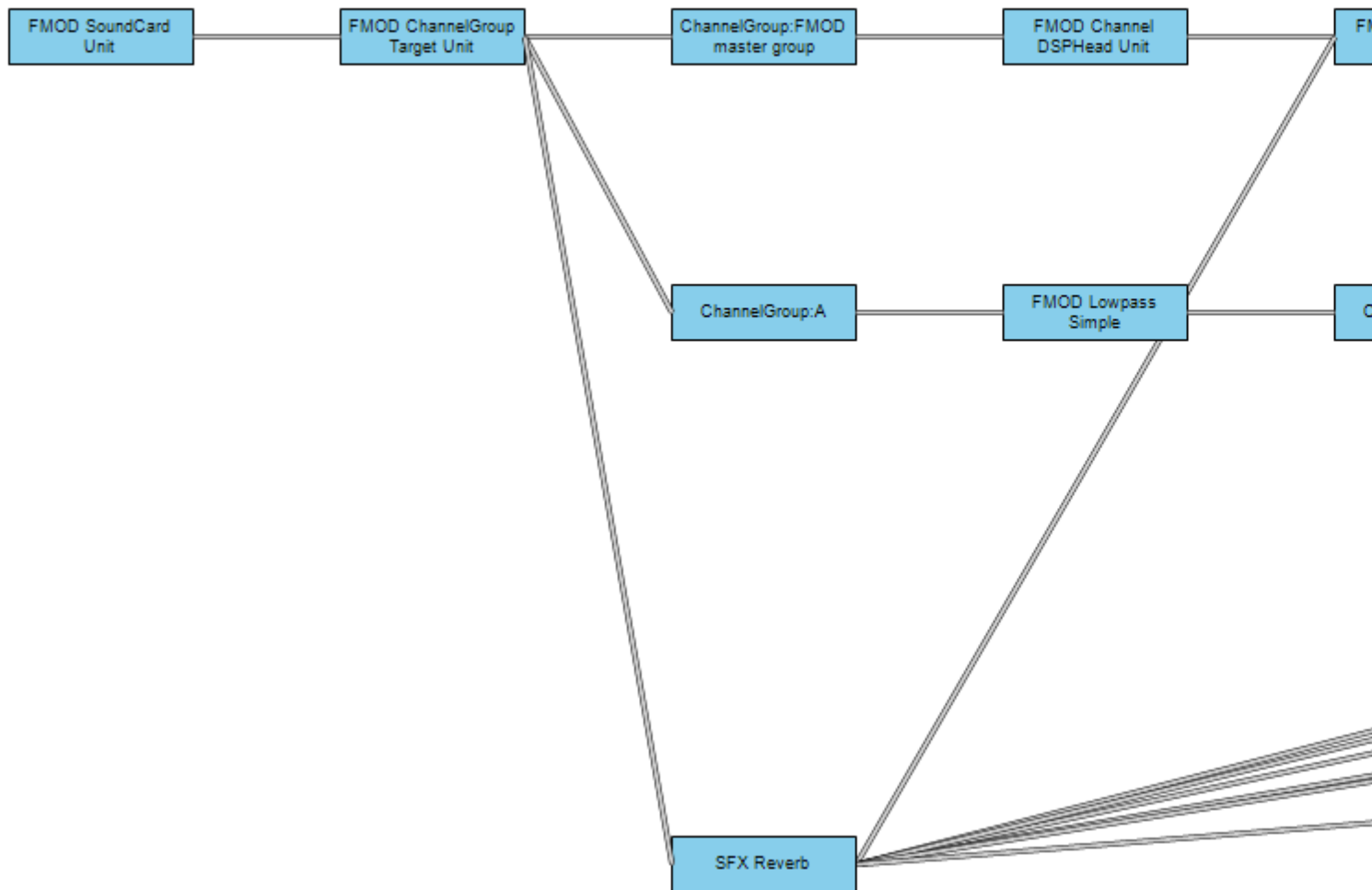
# DSP TUTORIAL

## Introduction

This section will introduce you to the FMOD Ex advanced DSP system. With this system you can do custom filters or complicated filter graph networks to create different and dynamic sounding audio.

The FMOD Ex DSP system is an ultimately flexible mixing engine, and goes far beyond FMOD 3's capabilities or any other audio mixing engine available right now.

Its emphasis on quality, flexibility and efficiency makes it an extremely powerful system if used to its full potential.



This is what an FMOD DSP network looks like. Audio data flows from the right to the left, until it finally arrives at the soundcard, fully mixed and processed.

The image was taken from a screenshot with the **FMOD DSPNet Listener** tool. You can run this on your own program as long as you specify [FMOD\\_INIT\\_ENABLE\\_DSPNET](#). The tool is located in the /tools directory of the SDK.

Some notes on this example image. Terms in **bold** are nodes that you can reference in the picture.

- When multiple inputs converge into one unit, they are mixed together. This is a submix.
- 7 channels are playing. These are depicted by **FMOD Channel DSPHead Unit**. Any time [System::playSound](#) or [System::playDSP](#) is called, an **FMOD Channel DSPHead Unit** is attached to the network. When it is stopped, it is disconnected from the network.
- To the right of each DSPHead Unit, there is either an **FMOD WaveTable Unit**, an **FMOD DSP Codec** or an



**FMOD Resampler Unit.** A WaveTable unit is a unit that plays standard PCM data (ie a standard wav file), a DSP Codec is a compressed realtime sample, created by [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) (in this case an mp3), and a Resampler Unit is a generic buffered resampler that is usually used when connecting generic DSP units to a channel (wavetable and DSPcodec usually have a built in resampler, standard DSP units do not so one has to be inserted if it is to be played on a channel, so [Channel::setFrequency](#) can work). These things feed data into a **Channel DSPHead Unit**, then it feeds the data onto its parent and so on.

- The 3 channels that have an **FMOD WaveTable Unit**, are playing a pcm wave file, and have been connected to a channelgroup. **ChannelGroup: B**. This was done with [Channel::setChannelGroup](#)
- The 3 channels that have an **FMOD DSP Codec Unit**, are playing an mp3 file and are connected to a channelgroup. **ChannelGroup A**. This was done with [Channel::setChannelGroup](#)
- The channel that has **UNIT A** attached to it is not attached to a user created ChannelGroup. It is connected to the default system ChannelGroup called the "master channelgroup" which is the ChannelGroup all channels are played on if no other is specified. You can get a handle to this channelgroup with [System::getMasterChannelGroup](#).
- The **UNIT A** unit was created by the user with [System::createDSP](#) and was played with [System::playDSP](#). It does nothing but act as a submix target, so the user has added 3 oscillators by creating them with [System::createDSPByType](#), and adding them to **UNIT A** with [DSP::addInput](#).
- You may notice that **ChannelGroup: B** is also connected to **ChannelGroup: A** as an input. This means the result of the submix of **ChannelGroup: B** is fed into **ChannelGroup: A**. The connection of B to A was done with [ChannelGroup::addGroup](#)
- ChannelGroup A and ChannelGroup B have 2 boxes each. This is because they both have a *DSP effect applied to them*. Without a DSP effect added, it just has one box, for optimization reasons (less memory, less cpu usage). The first box on the left is the head node, the second box on the right with the same name is the **mix target** for the channels. Inbetween is an **FMOD Echo DSP** ([FMOD\\_DSP\\_TYPE\\_ECHO](#)) effect on **ChannelGroup: B**, and a **FMOD Lowpass Simple** ([FMOD\\_DSP\\_TYPE\\_LOWPASS\\_SIMPLE](#)) effect on **ChannelGroup: A**. These were added with [ChannelGroup::addDSP](#)
- Reverb has been enabled with [System::setReverbProperties](#) and can be seen as a DSP node called **SFX Reverb**. Notice all channel based DSP units have a connection going to it. This is the *wet path*. The other paths go directly to the soundcard and bypass the SFX Reverb unit, therefore it is the *dry path*. To control the wet/dry mix you can use the [Channel::setReverbProperties](#). Internally this function just calls [DSP::setInputMix](#). Only channels are interested in this unit. Things like channelgroups and other units do not need to connect to this unit.
- If there was no reverb enabled, the secondary links/outputs (on the **FMOD WaveTable Unit/FMOD DSP Codec/FMOD Resampler Unit** units) would be absent.
- At the end of the mix, **ChannelGroup: FMOD master group**, **ChannelGroup: A** and **SFX Reverb** all get submixed into the **FMOD ChannelGroup Target Unit**
- Finally the result of that submix gets sent to the **FMOD SoundCard Unit** which is the final destination.
- If a DSP effect was to be added with [System::addDSP](#), it would be inserted between **FMOD SoundCard Unit** and **FMOD ChannelGroup Target Unit**. You could see why in that case it would affect all sound in the DSP network.
- If a DSP effect was to be added with [Channel::addDSP](#), it would be inserted to the right of the **FMOD Channel DSPHead Unit**. If this happened you should be able to see why only the channel would be affected by the DSP effect.

## Playing a sound and following the data flow.

When FMOD plays a *sound* on a channel (using [System::playSound](#)), it creates a small sub-network consisting of a **Channel DSP Head** and a **Wavetable Unit**.

When FMOD plays a *DSP* on a channel ([System::playDSP](#)), it creates a small sub-network consisting of a **Channel DSP Head** and a **Resampler Unit**. The DSP that was specified by the user is then attached to this as an input. This section will describe the units in more detail, from the origin of the data through to the soundcard, from right to left.

### **Wavetable Unit.**

This unit reads raw PCM data from the sound buffer and resamples it to the same rate as the soundcard. A

**Wavetable Unit** is only connected when the user calls [System::playSound](#).

After being resampled the audio data is then processing/flowing at the rate of the soundcard. This is 48khz by default.

### **Channel DSP Head.**

This unit does nothing. It simply is a place for extra DSP effects to connect to, between the **Wavetable Unit** (if [System::playSound](#) was used), or a user specified DSP unit (if [System::playDSP](#) was used), and the **ChannelGroup Unit** that it belongs to. By default all channels connect to the **Master Channel Group**.

It is also the unit where the channel volume and pan gets applied.

A Channel DSP Head unit incurs no CPU penalty. The data is simply passed straight to its outputs.

### **ChannelGroup DSP Heads.**

The **Master** [ChannelGroup](#) is the default target for **Channel DSP heads**, and is owned by the [System](#) object.

When multiple **Channel DSP Heads** are connected to a channel group, they are mixed together. This is the case for any DSP unit with multiple inputs.

Other channel groups may also be created by the user, which means channels may target them instead. This happens when the user calls [Channel::setChannelGroup](#)

Channelgroups are there for submixing. Effects can be placed after this point between it and the **ChannelGroup Target Unit**.

### **ChannelGroup Target Unit.**

This is the target DSP unit for all [ChannelGroup](#)s created by the user (with [System::createChannelGroup](#)) and the [System ChannelGroup](#).

# FMOD DESIGNER API PROGRAMMER'S TUTORIAL

## Introduction.

This section provides more technical information on how to use the FMOD designer API, and how resource allocation is handled to allow the programmer to account for performance and memory issues.

Just to provide some background information, the whole FMOD designer API sits on top of the low level FMOD API. This means it contains an FMOD::System object and uses all of the low level functions of the FMOD api to achieve its functionality.

## Files should you receive from the sound designer.

When the sound designer provides you with a project, they must provide you with the following files.

- 1 **.FEV** file. An FEV file is the compiled sound designer project which you will load with [EventSystem::load](#).
- 1 or more **.FSB** files. These files are raw audio data. They do not contain any event or sound designer data.
- Optionally, a project report ***project name.txt***. This is a file that describes the events to the programmer and any associated notes, along with the parameters for each event and their min/max values.
- Other files are working files (such as .cache), do not ship these.

## The programmer must work with the sound designer to organize banks and event groups to conserve memory!

Event groups should be used to control loading strategies, they are not just for aesthetic purposes, they are for loading purposes.

The branches of an event tree are what you use to load when in the game code.

See the "**Event tree group strategies and loading / memory allocation issues**" section below for very important issues related to loading and memory usage.

## Creating and initializing the EventSystem object.

Here is a typical bit of initialization code that you would call at the start of your project.

```
FMOD::EventSystem *eventSystem = 0;

result = FMOD::EventSystem::create(
    EVENT_SYSTEM_CALLBACK;

result = eventSystem->init(32, FMOD_INIT_NORMAL, 0);
EVENT_SYSTEM_CALLBACK;
```

If you want to configure the lower level FMOD engine before initializing the event system (ie select sound card driver,

set speaker mode etc), then call [EventSystem::getSystemObject](#).

```
FMOD::EventSystem *eventSystem = 0;
FMOD::System *system = 0;

result = FMOD::EventSystem::Create(
    &result, &system, &eventSystem, 0);
if (result != FMOD_OK)
    return;

result = eventSystem->getSystemObject(
    &result, &system, 0);
if (result != FMOD_OK)
    return;

// Use System API here from fmod.hpp.

result = eventSystem->init(32, FMOD_INIT_NORMAL, 0);
if (result != FMOD_OK)
    return;
```

**Do not create your own FMOD::System object** using [FMOD::System\\_Create](#).

This will cause 2 system objects to be active (the one you just created plus the one within the EventSystem), which means it will try to open the sound device twice. It also means 2 software mixers would be spawned. This is A Bad Thing.

Also do not try to create multiple EventSystem objects. If you want to load multiple projects, simply load them from the one EventSystem.

### **Important memory management issue for consoles (Xbox, Xbox 360, PlayStation2, GameCube, PlayStation Portable, PlayStation 3):**

Memory management is a consideration that must be taken note of. On certain machines the default memory allocation is inefficient (ie the page size is way too big on Xbox 360 meaning megabytes of lost memory), or certain FMOD features just won't work without a memory pool (ie Xbox must have one single block of contiguous memory for audio buffers, and on Xbox 360 XMA buffers MUST reside within memory allocated with XPhysicalAlloc, otherwise the system will crash).

Before calling any fmod functions, first allocate a block of memory and pass it to FMOD. From then on it will not allocate any more memory itself.

```
result = FMOD::Memory::Init(memBlock, MEMSIZE, 0, 0, 0);
if (result != FMOD_OK)
    return;
```

On Xbox 360 use XPhysicalAlloc.

Note that is the memory FMOD uses for all audio data including wave data, unless it is a machine with dedicated sound ram such as PlayStation 2 and GameCube.

In general it is usually a good idea to start off with a large memory block, then use [FMOD::Memory\\_GetStats](#) to find out the maximum memory usage by FMOD during the progress of the game.

[FMOD\\_ERR\\_MEMORY](#) will be returned from FMOD functions if it runs out of memory.

### **Virtual Voices.**

The value you pass to [EventSystem::Init](#) for the number of 'channels' should be a high number resembling the highest number of voices you want to have playing at once (not audible! there is a difference).

The hardware may only have 32 hardware voices, but this number can be 64, 128, 256 or 1000, because FMOD has a [Virtual Voice System](#).

If you do not give this a high number, then voice stealing will take effect, and voices will drop out seemingly at random.

That is because new, more important sounds will be played, kicking out older voices.

With a high enough number of virtual voices, no voices will be stolen, and FMOD will automatically swap voices in and out based on distance and priority. See the [virtual voice tutorial](#) in the tutorial for more information on this.

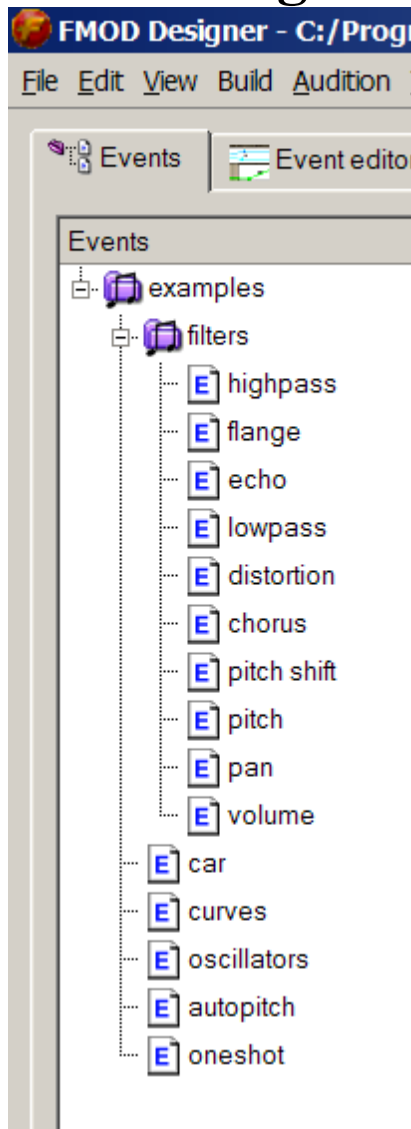
## Load the project.

Load the FEV file with [EventSystem::load](#). This only allocates memory for the event tree structure. It does not allocate memory for sample data / wave bank data or even the memory for the event instances, which are the things you use to play and control the events later in the code.

At this point the memory usage should be low and the memory allocated is for the low level software mixing engine and low level channel structures etc.

You can load multiple FEV files with this function into **EventProjects**.

## Traversing the event tree and getting events.



To traverse a tree you start at the root by calling [EventProject::getGroup](#) or [EventProject::getGroupByIndex](#) by specifying an event group at the root of the tree (ie 'examples' in this case).

From there you can call [EventGroup::getGroup](#) / [EventGroup::getGroupByIndex](#) to enter subgroups within the tree.

A full path can be entered within the GetGroup function, so that you don't have to manually traverse it yourself one at a time.

Finally you get a handle to an event with [EventGroup::getEvent](#) / [EventGroup::getEventByIndex](#).

These functions have loading (disk access) and memory allocation issues, which need to be considered in the next section.

# Event tree group strategies and loading / memory allocation issues. (Important!)

## Terminology.

**event** - The leaf node in the tree. The thing you will obtain a handle to so that you can play it (and update its parameters).

**event group** - 'Folders' that contain events and other event groups. These are used for organizational and loading purposes.

**wave banks** - The .FSB files to be loaded. When memory allocation occurs for these, it means allocation for the raw PCM or compressed audio data.

**event instance memory** - Memory required to play the event(s). If an event has a 'max playbacks' value set in the designer tool, FMOD will allocate memory for that many instances, so that they can play simultaneously. Generally the memory footprint for this is small unless the sound designer has specified memory intensive DSP effects such as reverb, echo, chorus or flange. Other types of DSP effects generally do not allocate any memory (IIR effects such as lowpass filter, distortion).

## Organization of event hierarchy and banks.

The event tree should be set up by grouping events into logical groups that will be loaded or used together, for example levels in a game, and common data.

The reason for this is that you can load an entire branch's audio data with [EventGroup::loadEventData](#). This data in particular is the (usually) larger **wave bank data**. This means it will load the waves from the **FSB files** referenced by the events in the tree. If it has already loaded sounds from the same FSB referenced from another event group, it will of course not try to re-load them.

**Note:** Sounds are either loaded selectively from an FSB, meaning you can pretty much put every sound into 1 wavebank, or you, as a programmer may prefer to preload a whole wavebank at once (ie [EventSystem::registerMemoryFSB](#)) which means you may want to split waves into logical wavebank groups.

## Loading / Allocation overview:

- [EventSystem::load](#) loads the FEV file, and only allocates a small amount of memory to hold the **event tree structure**.
- [EventGroup::loadEventData](#) loads all of the waves from the FSB files, necessary for the **specified group and its subgroups**. This function is recursive and traverses all subgroups.
- If you do not call [EventGroup::loadEventData](#), FMOD loads the event's wave data, when it needs to when you call [EventGroup::getEvent](#) / [EventGroup::getEventByIndex](#). If you do call [EventGroup::loadEventData](#) this won't happen. Calling the getevent function without loading the data first will mean a stall occurs as it loads. This is usually undesirable.
- [EventSystem::getGroup](#) / [EventGroup::getGroup](#) / [EventGroup::getGroupByIndex](#) allocate the **event instance memory** (including any DSP effect allocations) for the events in that group only, [EVENT\\_CACHEEVENTS](#) flag is used. This function is not recursive and does **NOT** traverse into subgroups.
- If **cacheevents = false** when getting a group, then FMOD will simply allocate the memory for the **event** when you try to get it. This means [EventGroup::getEvent](#) / [EventGroup::getEventByIndex](#) does the allocation.
- [EventGroup::freeEventData](#) unloads any wave data **AND** frees event instance memory for that group and all

groups below it. This function is recursive and **DOES** traverse into subgroups.

- [EventGroup::getEvent](#) / [EventGroup::getEventByIndex](#) will not do any disk access if [EventGroup::loadEventData](#) was called, or allocate any memory if you have precached it with a **GetGroup** function with **cacheevents = true**

#### **Just remember these things:**

1. Load your data at the loading phase of the game, with [EventGroup::loadEventData](#) from the root of a tree, generally for **static banks**.
2. Be selective when it comes to events using **streaming banks** that are in groups. If you only wanted 1 stream to play out of 10 in a group (ie all the tracks in your music group), then don't call [EventGroup::loadEventData](#) on the group, just call [EventGroup::getEvent](#) / [EventGroup::getEventByIndex](#) without having called [EventGroup::loadEventData](#) on the group, or using **cacheevents = true** on a **GetGroup** function.
3. [EventGroup::freeEventData](#) frees all memory related to a group and its children, including **wave bank** data and the **event instance memory**. If you call this then it will have to re-load and reallocate the data if you try to use that group again.

## **FMOD event system CPU usage.**

Now that the programmer has less control over the content of audio in the title, it may be easier to accidentally use more CPU than desired.

It is generally easy to find out FMOD cpu usage by using the low level functionality of FMOD with [EventSystem::getSystemObject](#) and [System::getCPUUsage](#). If the sound designer is using too many DSP effects, then the 'dsp' value will be high.

By default FMOD Designer puts all sounds in hardware, unless there is a DSP effect applied to the event. If there is then FMOD will load it into main ram, and mix the event on the CPU instead of in the audio chip.

Note all FMOD DSP effects are gradually being optimized using whatever SIMD capabilities are available on the machine. If an effect seems slower than it should be, it may possibly not be optimized, and request through [support@fmod.org](mailto:support@fmod.org) to have it done.

There are 12 effects and 11 platforms to optimize, which means 132 routines have to be optimized, and they also have to be optimized for mono/stereo and multichannel purposes so that is 396 loops to write, so as you can see the most important effects and platforms will be targeted first as we make our way through them.

# FMOD DESIGNER NETWORK API PROGRAMMER'S TUTORIAL

## Introduction.

This section provides instructions on how to use the FMOD Designer Network API. By using the FMOD Designer Network API, it is possible to use the FMOD Designer tool to connect to your game as it is running and tweak the properties of events as they're playing. This is useful for sound designers as they can, for example, play the game and adjust volume levels of events in realtime as they hear them.

## Using the NetEventSystem functions.

To use the FMOD Designer Network API all you need to do is :

- Call [NetEventSystem\\_Init](#) and pass it a pointer to your EventSystem object.
- Call [NetEventSystem\\_Update](#) just after each call you make to [EventSystem::update](#).
- Call [NetEventSystem\\_Shutdown](#) after you call [EventSystem::release](#).
- Link with `fmod_event_net.lib` as well as `fmod_event.lib`.

## Connecting to your game using FMOD Designer.

When your game is up and running using the FMOD Designer Network API, you can use the FMOD Designer tool to connect to it at any time.

To connect to your game :

- Select "Audition -> Manage Connections..." from the menu.
- Add a new connection and fill in the relevant details. You can use the loopback address 127.0.0.1 if you want to run your game and FMOD Designer on the same machine.
- Click "Connect".
- Load a project containing some or all of the events that will play in the game. Note: FMOD Designer can't trigger new events within your game. If you want to hear the results of your tweaking, you must trigger the relevant events from the game side.



- Now you can adjust the properties of event using the event property sheet in the event view, and envelopes / settings in event editor view.
- Save your changes in FMOD Designer using "File -> Save" at any time. Note: You must build the project again before these changes are made permanent in the .FEV file.

# FMOD Ex and movie players

## Introduction

This section describes how to have FMOD happily coexist with various movie playback systems available.

## PlayStation 2.

This section describes how to have FMOD happily coexist with various movie playback systems available. The main causes on conflicts between other middleware that uses audio or the IOP, and FMOD are:

**Conflict between the SPU2 DMA channels.** There are 2 of these. DMA channel 0 and DMA channel 1. By default FMOD uses SPU2 DMA channel 0 for software mixing, and DMA core 1 for uploading sample data and for streaming to. This means DMA core 1 is used when [System::createSound](#) is being executed to load a PS2 FSB file, or streaming using [System::createStream](#). To work around this issue see the following tips.

- **Turn off the FMOD software mixer.** This is already done if you are using the \_reduced version of the library. This will free up DMA Channel 0. Most of the time you are not going to need the FMOD software mixer. You can do this by using [System::init](#) with the [FMOD\\_INIT\\_DISABLESOFTWARE](#) flag.
- **Swap FMOD's mixer/upload channel usage around.** If the 3rd party software still uses DMA Channel 1 (the channel FMOD uses for bank uploads and streaming), you can either change your 3rd party software to use DMA Channel 0 instead of DMA Channel 1, or tell FMOD to swap its usage around by specifying [FMOD\\_INIT\\_PS2\\_SWAPDMACHANNELS](#). If you didn't turn the software mixer off, this would make FMOD use DMA Channel 0 for streams and sample bank uploads, and DMA Channel 1 for the software mixer.

**Conflict on the SIFCMD ports.** If your middleware or your own code is using the SIFCMD sony library to communicate with the IOP, then if you don't take care to share the SIFCMD buffers and ports with FMOD, messages will get lost and unexpected behaviour will occur in FMOD and your 3rd party software.

- **If you want to initialize your 3rd party software after FMOD.** Use this information if the code has a way to set up its SIFCMD usage. Note that FMOD uses SIFCMD port 0 and 1, and has a **buffer size of 16**.
- **If you want to initialize your 3rd party software before FMOD.** If FMOD is initialized first, call [System::init](#) and use FMOD\_PS2\_EXTRADRIVERDATA structure from **fmodps2.h**. Also load your FMODEX.IRX or FMODEXD.IRX with command line parameters to allow the IOP side to get the same information. For more detailed information on this see the comment above the FMOD\_PS2\_EXTRADRIVERDATA declaration in **fmodps2.h**.

**SPU2 ram usage and SPU2 hardware voice usage.** Because FMOD and the middleware might not know about each other, they might allocate memory or use SPU2 voices without any regard for the other.

- **Use FMOD\_SPU2\_Alloc / FMOD\_SPU2\_GetRawAddress / FMOD\_SPU2\_Free.** Use these functions to set aside SPU2 ram for other middleware usage.
- **Use FMOD\_SPU2\_ReserveVoice.** Use this function to set aside an SPU2 voice for other middleware usage.

## Xbox 1

The main causes on conflicts between other middleware that uses audio on the XBox, and FMOD are:

**"DSOUND: CMcpxApu::AllocateVoices: Error: Not enough free hardware voices".** By default FMOD assumes it is in total control of the audio, so it allocates every XBox audio voice. If another 3rd party software application tries to allocate a hardware voice it will fail.

- Use [System::setHardwareChannels](#). To get around this issue just call [System::setHardwareChannels](#) to reduce the count. XBox has around 192 HW2D voices so you could reduce this and still have plenty of voices free.
- **DSP Image incorrect.** FMOD has an internal MCP DSP image that it loads. It is stripped down to save memory, (hundreds of kilobytes) by removing unnecessary features. This may conflict with other 3rd party software that relies on a standard DSP image such as Microsoft's dsstdfx.bin.
- **Use FMOD\_SpecifyEffectsImage.** This functionality is not available at this time. Contact support.
- **Make the other software use FMOD's image.** This functionality is not available at this time. Contact support.
- **Needs access to the XBox LPDIRECTSOUND handle.** If the 3rd party software is initialized second, it may want to use the XBox LPDIRECTSOUND handle.
- Use [System::getOutputHandle](#). If you need to get a handle to FMOD's internal DirectSound pointer, you can share it by calling [System::getOutputHandle](#) and casting it to the appropriate pointer type.

## Windows

If you need to get a handle to FMOD's internal DirectSound pointer, you can share it by calling [System::getOutputHandle](#) and casting it to the appropriate pointer type. This also goes for XBox and other platforms.

## Bink on PS2

Here is a quick way to get FMOD PS2 and Bink to co-exist.

Initialize FMOD first.

```
FMOD_System_t(system, mmchans, FMOD_INIT_32BIT_SPEAKER, NULL);
```

Initialize Bink

```
Movie_Buffer = sceSiAllocIoPHeader(ADIO_MemoryAmount(ADIO_PRIMARY));
ADIO_MemoryAmount(ADIO_PRIMARY);
if (!ADIO_BartUp(1, 2, Movie_Buffer, ADIO_PRIMARY | ADIO_PRIMARY_INTERRUPT))
{
    sceSiFreeIoPHeader(Movie_Buffer);
    // error
}
ADIO_PrimaryVolumes(1, 0xffff, 0xffff, 0xffff, 0xffff);
BinkSoundUseADIO_P(1);
```

# THREADS AND THREAD SAFETY

## Introduction

This section will talk about the threads FMOD creates, and thread safety.

## FMOD threads types.

FMOD has 4 main threads, and 2 of which are created at the time of [System::init](#), and 2 of which are created only when you use certain flags.

They are

- **Mixer thread.** Software mixing thread, created at [System::init](#).
- **Stream thread.** Thread used for decoding streams. Created the first time a sound is loaded as a stream in [System::createSound](#) with [FMOD\\_CREATESTREAM](#) or [System::createStream](#).
- **Async loading thread.** Created the first time a sound is loaded with the [FMOD\\_NONBLOCKING](#) flag in [System::createSound](#).
- **File reading thread.** Thread used for reading from disk for streams, to then be decoded (decompressed) by the Stream thread. Created the first time a sound is loaded as a stream in [System::createSound](#) with [FMOD\\_CREATESTREAM](#) or [System::createStream](#).

### Exceptions.

If [FMOD\\_INIT\\_STREAM\\_FROM\\_UPDATE](#) is used, then the stream thread will not be created.

If [FMOD\\_OUTPUTTYPE\\_WAVWRITER\\_NRT](#) or [FMOD\\_OUTPUTTYPE\\_NOSOUND\\_NRT](#) are used, then the mixer thread will not be created.

Everything else is run from the **main / game** thread, including [System::update](#) calculations.

## FMOD threads priorities per platform.

For reference, for your own thread code, here are the thread priorities used in FMOD per platform.

Thread priority table.

Platform	Mixer thread	Stream thread	Async loading thread	File reading thread
Win32/Win64 (SetThreadPriority)	THREAD_PRIORITY_TIME_CRITICAL	THREAD_PRIORITY_HIGHEST	THREAD_PRIORITY_ABOVE_NORMAL	THREAD_PRIORITY_ABOVE_NORMAL
Linux/Linux64 (pthread_setschedparam)	99	94	90	90
Mac/Mac86 (MPSetTaskWeight)	10000	9000	8000	8000

PS2 (EE, CreateThread)	1	10	17	17
PSP (sceKernelCreateThread)	8	12	16	16
PS3 (sys_ppu_thread_create)	0	300	600	600
Xbox (SetThreadPriority)	THREAD_PRIORITY_TIME_CRITICAL	THREAD_PRIORITY_HIGHEST	THREAD_PRIORITY_ABOVE_NORMAL	THREAD_PRIORITY_ABOVE_NORMAL
Xbox 360 (HW Thread 4, SetThreadPriority)	THREAD_PRIORITY_TIME_CRITICAL	THREAD_PRIORITY_HIGHEST	THREAD_PRIORITY_ABOVE_NORMAL	THREAD_PRIORITY_ABOVE_NORMAL
Gamecube (OSCreateThread)	0	8	12	12
Wii (OSCreateThread)	0	8	12	12

## FMOD callbacks

FMOD File and memory callbacks can possibly be called from an FMOD thread. Remember that if you specify file or memory callbacks with fmod, to make sure that they are thread safe. FMOD may call these callbacks from the stream thread, or [FMOD\\_NONBLOCKING](#) thread at any time.

## Calling FMOD commands from different threads.

Do not call FMOD commands from different threads! This will lead to instability, corruption and possible crashes.

Some people are tempted to put [System:update](#) into a separate thread, **Only do this if you criticalsection this and all other calls to fmod..**

To make FMOD thread safe would involve wrapping every FMOD function in a critical section, which adds unnecessary overhead, so at this time FMOD Ex is remaining 'not thread safe' for the time being.

# REVERB NOTES

## Introduction

This section will discuss FMOD's reverb parameters. For a more general description of reverb, see [here](#).

The fields of FMOD\_REVERB\_PROPERTIES (found in 'fmod.h') control both hardware (via EAX) and software (via SFX) instances of reverb. EAX has a few parameters that the software doesn't use, so you can ignore these for the purpose of this discussion. For example, EnvSize is NOT used by the software reverb because it is only meaningful to EAX.

FMOD's software reverb DSP is controlled by parameters defined in the [I3DL2 guidelines](#), which describe the reverberant environment of the listener.

Here's a list of the fields of FMOD\_REVERB\_PROPERTIES that currently have an effect and a description of what they do within the context of the software reverb. The descriptions are much the same as in 'fmod.h'.

<b>Environment</b>	Turns the reverb off if set to -1
<b>Room</b>	Room effect level (at mid frequencies)
<b>RoomHF</b>	Relative room effect level at high frequencies
<b>DecayTime</b>	Reverberation decay time at mid frequencies
<b>DecayHFRatio</b>	High-frequency to mid-frequency decay time ratio
<b>Reflections</b>	Early reflections level relative to room effect
<b>Reverb</b>	Late reverberation level relative to room effect
<b>ReverbDelay</b>	Late reverberation delay time relative to initial reflection
<b>Diffusion</b>	Echo density in the late reverberation decay
<b>Density</b>	Modal density in the late reverberation decay
<b>HFRference</b>	Reference high frequency (Hz) [see RoomHF]
<b>RoomLF</b>	Relative room effect level at low frequencies
<b>LFReference</b>	Reference low frequency (Hz) [see RoomLF]
<b>Room</b>	Room effect level (at mid frequencies)

## Please note :

RoomRolloffFactor is a part of I3DL2, but has no effect within FMOD.

As with most reverberation models, the response is split into sections. This implementation has early reflections and late reverberation, each of which are composed of sets of delay lines having different delay and decay characteristics.

There are a few things to note here:

- 1) Room, RoomHF, RoomLF, Reflections and Reverb are all measured in milliBels, i.e. 100th of a deciBel, and they're all integers.
- 2) Room is the input gain
- 3) Reflections is a gain on the output of the early reflections subsystem
- 4) Reverb is a gain on the output of the late reverb subsystem

- 5) RoomLF and LFReference control a low frequency shelving filter on the input
- 6) RoomHF and HFReference control a high frequency shelving filter on the input
- 7) Diffusion and Density control the correlation among delay lines in the reverb subsystem

The FMOD\_PRESET\_\* presets can be useful as examples of how these parameters change the nature of the reverb. They give quite a wide scope for representing different environments.

One more thing - it's important to distinguish between FMOD\_REVERB\_PROPERTIES and FMOD\_REVERB\_CHANNELPROPERTIES. The latter is just used for controlling the a channel's input gain to the reverb, and doesn't affect the characteristics of the reverb unit itself.

# WINDOWS SPECIFIC ISSUES / FEATURES

## Installation

- Use **api/fmodex.dll** to use FMOD Ex with all plugins statically compiled into the DLL. This means you can use all the features of FMOD without needing extra plugins accompanying your application. The DLL is bigger because of this.
- Use **api/fmodexp.dll** to use FMOD Ex with plugins external. This DLL needs plugins to function, which you can find in the plugins directory. Plugins in the plugins/ directory need to be used to support all of FMOD Ex's features. Use this if you want a smaller distribution and only need one file format support for example (ie .WAV). The DLL is smaller because of this.

## Linking (which library to link to)

If you want to use fmodex.dll: (all plugins compiled into the dll, larger main dll size)

- Visual Studio users - **fmodex\_vc.lib**.
- Metrowerks Codewarrior users - **fmodex\_vc.lib**.
- Intel compiler users - **fmodex\_vc.lib**.
- Borland users - **fmodex\_bc.lib**.
- LCC-Win32 users - **fmodex\_lcc.lib**.
- Dev-C++, MinGW and CygWin users - **libfmodex.a**.

If you want to use fmodexp.dll: (plugins left external, smaller main dll size).

- Visual Studio users - **fmodexp\_vc.lib**.
- Metrowerks Codewarrior users - **fmodexp\_vc.lib**.
- Intel compiler users - **fmodexp\_vc.lib**.
- Borland users - **fmodexp\_bc.lib**.
- LCC-Win32 users - **fmodexp\_lcc.lib**.
- Dev-C++, MinGW and CygWin users - **libfmodexp.a**.

## Recommended start up sequence (IMPORTANT!).

Due to configuration issues on Windows user's machines, this following code fixes the following issues:

- Speaker configuration in windows being ignored and just defaulting to stereo. (see use of [System::getDriverCaps](#) and 'controlpanelspeakermode' parameter, which is then passed to [System::setSpeakerMode](#))
- Stuttering audio due to the user having their 'Hardware acceleration' slide set to 'off' in XP. (see check for [FMOD\\_CAPS\\_HARDWARE\\_EMULATED](#), which then increases the FMOD DSP buffersize to over 200ms with [System::setDSPBufferSize](#))
- Speaker configuration being set to a setting that the soundcard *doesn't actually support* (See check for



[FMOD\\_ERR\\_OUTPUT\\_CREATEBUFFER](#), which then triggers a reinitialization with [FMOD\\_SPEAKERMODE\\_STEREO](#))

**!!! THIS CODE MUST BE USED FOR SHIPPING GAMES. DO NOT SHIP A GAME WITHOUT A STARTUP SEQUENCE BASED ON THIS CODE !!!**

Use the following code as a basis for your Windows start up sequence.

```
FMOD_System sys;
FMOD_RESULT result;
unsigned int channels;
FMOD_SPEAKERMODE speakermode;
FMOD_Caps caps;

/*
    Create a System object and initialize.
*/
result = FMOD_System_Create(&sys);
ERR_CHECK(result);

result = sys->GetChannels(&channels);
ERR_CHECK(result);

if(channels < 2)
ERR_CHECK(result);

result = sys->SetSpeakerMode(speakermode); /* Set the user selected speaker mode. */
ERR_CHECK(result);

if(caps & FMOD_CAPS_48KHZ) /* You might want to warn the user about this. */
    result = sys->SetFrequency(48000); /* At 48khz, the delay between issuing a command and hearing it will now be about 213ms. */
ERR_CHECK(result);

result = sys->SetMinMaxChannels(100, FMOD_MIN_CHANNELS, 0); /* Replace with whatever channel count and flags you use! */
if(result == FMOD_ERR_OUTPUT_CREATEBUFFER) /* Ok, the speaker mode selected is unsupported by this soundcard. Switch it back to stereo... */
{
    result = sys->SetSpeakerMode(FMOD_SPEAKERMODE_STEREO);
ERR_CHECK(result);

    result = sys->SetMinMaxChannels(100, FMOD_MIN_CHANNELS, 0); /* Replace with whatever channel count and flags you use! */
ERR_CHECK(result);
}
```

**Important issue with Borland, LCC-Win32, Dev-C++, MinGW, Cygwin users and FMOD Ex C++ interface.**

Note that due to incompatible linking standards with C++ symbols in libraries across different compilers, you will not be able to use the C++ interface of FMOD Ex with these compilers.

You can only use the FMOD Ex C interface, as at least that has a compatible standard (ie stdcall symbols are always the same format).

Each C++ compiler generates its own version of mangled symbols, and the mentioned compilers are not compatible with the symbols that MSVC produces, which is what FMOD is compiled in, and is the more popular compiler for commercial development at this stage.

Note that the Intel compiler and Codewarrior do not have this problem, they can resolve MSVC style symbols.

## Troubleshooting.

### Stuttering/skipping sound when using software mixed sounds, or streams.

More commonly known as buffer underrun/overflow, this can be 1 or a combination of factors:

- **Bad soundcard drivers** - This may be solved by upgrading your soundcard drivers. (Note it is recommended you get the latest drivers anyway)
- **CPU issues** - Machine too slow, or whatever you are trying to do with FMOD is too CPU intensive! (ie playing 100 mp3's at once will most likely bring FMOD to its knees, or maybe a user stream callback or DSP callback is spending too much time executing).
- **Mixer buffersize is set too small** - You can increase stability to combat these issues, by increasing FMOD's internal mixing buffer size. This will lead to greater stability but also larger latency on issuing commands to hearing the result. Call [System::setDSPBufferSize](#) to alter this. See documentation for [System::setDSPBufferSize](#) for more information.
- **Stream buffersize is set too small** - If you are using the FMOD Ex streamer, you might be streaming from a slow media, such as CDROM or over network, or even a fragmented harddisk, therefore FMOD needs more time to fill its streaming buffer before it runs out. See [System::setStreamBufferSize](#) to adjust the file read buffer size for the streamer. If the stream is starving because the codec is an expensive codec (and the file media is not to blame) then the problem could be the FMOD stream decode buffer size. You can adjust this using the 'decodebuffersize' member of the [FMOD\\_CREATEINDEXINFO](#) structure.
- **Output type** - [FMOD\\_OUTPUTTYPE\\_DSOUND](#) will provide more solid output than [FMOD\\_OUTPUTTYPE\\_WINMM](#) in anything except Windows NT. This is a problem with Windows Multimedia Services not being as realtime as it should be. Under NT [FMOD\\_OUTPUTTYPE\\_WINMM](#) is more stable, as DirectSound in NT is just emulated by using WINMM itself and is actually slower and has longer latency!. **Note!** Please don't feel the need to use [System::setOutput](#) if you don't need to. FMOD autodetects the best output mode based on the operating system.

# LINUX SPECIFIC ISSUES / FEATURES

## Installation

- Use **api/lib/libfmodex.so** to use FMOD Ex with all plugins statically compiled into the library. This means you can use all the features of FMOD without needing extra plugins accompanying your application. The library is bigger because of this.
- Use **api/lib/libfmodexp.so** to use FMOD Ex with plugins external. This library needs plugins to function, which you can find in the plugins directory. Plugins in the plugins/ directory need to be used to support all of FMOD Ex's features. Use this if you want a smaller distribution and only need one file format support for example (ie .WAV). The library is smaller because of this.

## Formats not supported.

WMA is the only file format not support on FMOD Ex for linux. This is because FMOD uses a windows codec to be able to decode WMA. This codec is proprietary and owned by Microsoft and is not cross platform.

# MAC SPECIFIC ISSUES / FEATURES

## Installation

- Use **api/lib/libfmodex.dylib** to use FMOD Ex with all plugins statically compiled into the library. This means you can use all the features of FMOD without needing extra plugins accompanying your application. The library is bigger because of this.
- Use **api/lib/libfmodexp.dylib** to use FMOD Ex with plugins external. This library needs plugins to function, which you can find in the plugins directory. Plugins in the plugins/ directory need to be used to support all of FMOD Ex's features. Use this if you want a smaller distribution and only need one file format support for example (ie .WAV). The library is smaller because of this.

## Universal Binaries

All FMOD Ex libraries for Macintosh are shipped as universal binaries, this means FMOD will support applications designed for either PowerPC, x86 or both. This causes the libraries to be around double the size compared with those that only support one variant. If you are only targeting one platform, either PowerPC or x86 you can extract smaller individual libraries from the provided larger one using the "**lipo**" tool.

ie.

```
i p - th n pp i bmo d x. q i b-ou tp t i bmo d x pp . q i b
```

o r

```
i p - th ni3 86 i bmo d x. q i b-ou tp t i bmo d x x86. q i b
```

# PLAYSTATION 2 SPECIFIC ISSUES / FEATURES

## Installation.

### EE libraries.

Link this into your project. One of these files must be linked.

#### GCC / ProDG users.

- **/api/lib/fmodex.a** for general development with all possible features included. Software mixing features will incur higher CPU usage.
- **/api/lib/fmodexD.a** for the same library, but with debug logging which can help to determine any problems if they exist.
- **/api/lib/fmodex\_reduced.a** for general development with a smaller library size and features removed. See table below for which features are removed. Recommended for PS2, as it lowest memory usage and cpu usage.
- **/api/lib/fmodexD\_reduced.a** for the same library, but with debug logging which can help to determine any problems if they exist.

#### Codewarrior users.

- **/api/lib/fmodex\_cw.lib** for general development with all possible features included. Software mixing features will incur higher CPU usage.
- **/api/lib/fmodexD\_cw.lib** for the same library, but with debug logging which can help to determine any problems if they exist.
- **/api/lib/fmodex\_reduced\_cw.lib** for general development with a smaller library size and features removed. See table below for which features are removed. Recommended for PS2, as it lowest memory usage and cpu usage.
- **/api/lib/fmodexD\_reduced\_cw.lib** for the same library, but with debug logging which can help to determine any problems if they exist.

### IOP module.

Load this into your project at runtime. You have to load the IRX yourself using sceSifLoadModule. More about this follows.

- **/api/fmodex.irx** for general development.
- **/api/fmodexD.irx** for the same IRX, but with debug logging which can help to determine any problems if they exist.

Feature table.

Feature	fmodex.a	fmodex_reduced.a	Requires software mixing?
Streaming audio support	Y	Y	N
3D Sound	Y	Y	N
Virtual voices	Y	Y	N
FMOD Designer API support	Y	Y	N
Nonblocking sound open support	Y	Y	N

Hardware reverb	Y	Y	N
Geometry support / polygon occlusion	Y	N	N
Software mixing	Y	N	Y
Spectrum Analysis	Y	N	Y
Network streaming	N	N	n/a
Recording support	N	N	n/a
File format - FSB	Y	Y	N
File format - VAG	Y	Y	N
File format - AIFF	Y	N	Y
File format - DLS	Y	N	Y
File format - FLAC	Y	N	Y
File format - IT (sequenced music format)	Y	N	Y
File format - MIDI (sequenced music format)	Y	N	Y
File format - MOD (sequenced music format)	Y	N	Y
File format - MP2 / MP3	Y	N	Y
File format - Ogg Vorbis	Y	N	Y
File format - M3U / PLS / ASX (Playlist format)	Y	N	Y
File format - RAW (format specified by user)	Y	N	Y
File format - S3M (sequenced music format)	Y	N	Y
File format - Tag formats - ID3V2, ASF, Ogg tags	Y	N	N
File format - XM (sequenced music format)	Y	N	Y
File format - WAV	Y	N	Y
File format - User created	Y	Y	N
File format - ASF / WMA	N	N	n/a
File format - CDDA	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_PS2	Y	Y	N
Output mode - FMOD_OUTPUTTYPE_WAVWRITER	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_WAVWRITER_NRT	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_NOSOUND	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_NOSOUND_NRT	N	N	n/a
DSP Filter - Oscillator	Y	N	Y
DSP Effect - Lowpass	Y	N	Y
DSP Effect - Lowpass2	Y	N	Y
DSP Effect - Highpass	Y	N	Y
DSP Effect - Echo	Y	N	Y
DSP Effect - Flange	Y	N	Y
DSP Effect - Distortion	Y	N	Y
DSP Effect - Normalize	Y	N	Y
DSP Effect - Parameq	Y	N	Y
DSP Effect - Pitchshift	Y	N	Y
DSP Effect - Chorus	Y	N	Y
DSP Effect - Software reverb	Y	N	Y
DSP Effect - IT echo	Y	N	Y
DSP Effect - SFX reverb	N	N	n/a

In this table "**Requires software mixing?**" is specified to let the user know that the main CPU and RAM will be used to perform the feature which may not be desirable for the programmer.

Most things requiring the FMOD software mixer are removed in the reduced version of the library, to provide simple sound support.

The FMOD designer API can be used with the reduced library as well as long as all banks are marked as hardware in FMOD designer.

Note with a source code license you can easily turn features on and off to reduce code size or create different combinations of features to best suit your needs.

## Loading IRIX modules required for FMOD to operate.

Put **/api/fmodex.irm** into your modules directory. You have to load the IRIX yourself using sceSifLoadModule.

A simple PlayStation 2 application has to do the following to use FMOD.

1.

Load **fmodex.irm** and Sony's **libsd.irm**. This is done from host0 or cdrom0 or whatever file device you store your files on.

Load libsd first, then fmod.

eg.

```
wh_e (sceSifba_0_0_e (hs_t0mo_0_0_s/ i_b_d_i_r_x", 0, NULL
```

N\_e th\_t th\_e d\_au\_l\_t\_psi\_tio\_n\_f\_r\_i\_b\_d\_i\_r\_xis a\_t/us\_r/bca\_l/sce/io\_p/mo\_0\_0\_s/ b\_t\_i\_t may\_d\_f\_e\_r\_o\_n\_yo\_u\_rmac\_h\_e.

**W\_h\_a\_t\_doe\_s\_eac\_hmo\_0\_0\_e\_d?**

- i\_b\_d\_i\_r\_x - This is a so\_y\_i\_r\_x th\_t co\_n\_tai\_n\_s th\_e b\_w\_e\_l\_h\_r\_d\_a\_e\_r\_o\_u\_t\_i\_n\_s a\_e\_d\_d\_b\_y\_m\_o\_d\_x\_i\_r\_x.
- m\_o\_d\_x\_i\_r\_x - This is th\_e m\_o\_d\_i\_b\_a\_y\_a\_n\_d co\_n\_tai\_n\_s th\_e m\_a\_j\_o\_r\_i\_t\_y\_o\_f th\_e f\_u\_n\_c\_t\_i\_o\_n\_s.

2.

R\_e\_m\_e\_m\_b\_e\_r\_t\_o\_i\_n\_i\_t\_i\_a\_l\_e th\_e IO\_P\_h\_a\_p\_w\_i\_th\_sceSif\_n\_t\_o\_p\_h\_a\_p().

This is t\_o\_b\_e\_c\_a\_l\_l\_e\_d\_a\_f\_t\_e\_r th\_e c\_a\_l\_l\_t\_o\_sceSif\_n\_t\_Rp() a\_n\_d\_b\_e\_f\_o\_r\_e b\_a\_d\_g\_a\_y\_m\_o\_d\_0\_0\_s.

eg.

```
sceSif_n_tRp(0;
sceSif_n_t_o_p_h_a_p();
WA_R_N_I I_f_yo_u_e_b_o_t_th_e IO_P_yo_u_h_v_e_b_e_c_a_l_l_th_e_a_b_v_e_a_g_a_i_n!! O_t_h_w_i_s_e_FMO_D_w_i_l_l
f_a_i_l_t_o_i_n_i_t_i_a_l_e.
```

3 .  
Note that mismatching mode and mode extensions are not allowed.  
MODE will fail to initialize if they are from different bases of MOD.

## Co dwa rio ruse s ca llmwI n tð!

MODE is a C++ library and needs to be installed. Make sure you call mwI n t a t t h s t a r t o f y o u r p r o g r a m .  
If you don't do this you will get errors a most time due to from [MODE: Sys tem C re a t e](#) .

## S U 2Sou nd am .

The Phys ics 2 has 2Bosound am which can be compressed VAG data (3.5:1 compression) DEM and it, but you cannot access all 2mb.  
MODE uses most of the am, but some is set aside for hardware work areas.

Input/Output Area (21k)	VAG Sound data (1515k)	CORE0 Reverb work area (256k)	CORE1 Reverb work area (256k)
-------------------------	------------------------	-------------------------------	-------------------------------

The first 2k is the hardware work area which cannot be used to store sound data.

If you don't plan to use reverb, you can gain an additional 1k of the reverb work area back for sound data.  
ie .

Input/Output Area (21k)	VAG Sound data (2027k)
-------------------------	------------------------

You can do this by specifying [MODE\\_INT\\_B2\\_DSA BECO RE VERB](#) | [MODE\\_INT\\_B2\\_DSA BECO RE VERB](#) in the flags region of [Sys tem : i n t](#) .

CORE0 reverb affects hardware voices 0 to 3, and CORE1 reverb affects hardware voices 24 to 7. Currently there is no way to make MOD play a sound on a particular core but this will be a default in the future.

## EE Th ead P r o i t y .

Note that the EE main thread priority is not changed from the default of 1, MOD will change it to 32.  
The thread can start when the default priority is 1 so this is necessary.  
MODE uses 1 thread to receive messages from the IOP.

## B n n g MO DE am p e s .

All the ampers are for files that are shared to the elf. In the get manager for the ampers, tick the 'Set file serving port' to make host the same directory as the elf.

## The B B f m a t - The recommended format for samples and teams .

Although .WAV and .VAG formats are supported, for better speed and teaming speed it is highly recommended to use .B B f i l e s .  
B B is hardware accelerated, WA Vis not.



It is composed of a PHY Station 2 SUR2 sound data, a range of when loaded, it is loaded to the data headers first, then the raw data (which is stored continuously), and when loaded it is efficiently streamed into SUR2 ram. This is the fastest way to load sound data.

## PHY Station 2 hardware error.

You have access to the hardware PHY Station 2 SUR2 error through MODs errorAE. Note that the SUR2 Error is a bit more primitive than I3 DL2 error and EAS. In the MOD\_ERROR\_PROPERTIES structure, only Environment, Rom and Flags are supported.

### 'Environment'

This is a value between 0 and 9 mapping to the sound error modes. You will find 9 special cases for the PHY Station 2 with this environment value set accordingly.

ie

```
MOD_ERROR_BIT_2_ROM
MOD_ERROR_BIT_2_STUDIO_A
MOD_ERROR_BIT_2_STUDIO_B
MOD_ERROR_BIT_2_STUDIO_C
MOD_ERROR_BIT_2_HALL
MOD_ERROR_BIT_2_STAGE
MOD_ERROR_BIT_2_EC0
MOD_ERROR_BIT_2_E1A
MOD_ERROR_BIT_2_EE
```

The other cases will not work, except for MOD\_ERROR\_TOFF.

### 'Rom'

This still controls the amount of error bits in the output. Normally it is in decibels, between -10000 (silent) and 0 (full volume), and it is the same range on the PHY Station 2, but it is a linear scale between -10000 and 0, not a logarithmic one.

### 'Flags'

This only utilizes the following fields on PHY Station 2.

```
MOD_ERROR_FLAGS_COE0 (hardware vices 0 to 3)
MOD_ERROR_FLAGS_COE1 (hardware vices 24 to 4)
```

This tells the MOD Engine which cores, or set of hardware vices to apply the error settings to.

By default (in the cases) it is set to apply to both cores, but you can remove these flags to control each core separately.

Note that [Channel:serverProperties](#) is supported through the 'Rom' parameter only, and that this value is binary, ie -10000 is 'error off' for the channel, and anything else is 'error on'.

# PLAYSTATION 3 SPECIFIC ISSUES / FEATURES

## Installation.

FMOD libraries were built using PS3 SDK 210.001.

### PPU libraries.

Link this into your project. One of these files must be linked.

- Use **/api/lib/fmodex.a** for general development with all possible features included.
- Use **/api/lib/fmodexL.a** for the same library, but with debug logging which can help to determine any problems if they exist.

The following libraries should also be linked into your project:

- libm.a
- libfs\_stub.a
- libspurs\_stub.a
- libsysutil\_stub.a
- libaudio\_stub.a

## SPU Threads and SPU Threads priorities.

When using SPU Threads, the SPU version of FMOD requires the SPU modules:

- **/api/fmodex\_spu.self**
- **/api/fmodex\_spu\_mpeg.self**

The path to the SPU modules must be passed in through the **FMOD\_PS3\_EXTRADRIVERDATA** structure declared in **fmodps3.h**.

If you wish to load the SPU modules from memory, pass in pointers rather than the file paths and set the **spu\_load\_from\_memory** member to 1.

The SPU priority of the mixer, mpeg stream decoder and at3 decoder is also specified in this structure. For more information on the **FMOD\_PS3\_EXTRADRIVERDATA**, please refer to the **fmodps3.h** header.

A pointer to this structure should be passed in as **extradriverdata** to [System::init](#).

ie.

```
FMOD_PS3_EXTRADRIVERDATA extradriverdata;

memset(

    extradriverdata.spu_mixer_selfname_or_spu_data, 0,
    SYS_APP_NAME_LEN * sizeof(char));
```

```

    e x t a d i t e r d a s p _ s t r e a m e r _ e l f n a m e _ o r s p u r s d a t a      =
SYS_A P P _ D M E "/" f m o d _ x s p _ m p e g _ s e l f";

    e x t a d i t e r d a s p _ b a d _ f o m _ m e m o r y                                = 0      /* S e t t h i s t o 1 i f w e
h a v e p a s s e d i n p i n t e r s t o e m b e d d e d l f d a t a */

    e x t a d i t e r d a s p _ p r i o r i t y _ m i x e r                                = 16      /* D e f a u l t */
    e x t a d i t e r d a s p _ p r i o r i t y _ a t t e m p t s                         = 200     /* D e f a u l t */
    e x t a d i t e r d a s p _ p r i o r i t y _ s t r e a m e r                        = 200     /* D e f a u l t */

    e x t a d i t e r d a s p u r s = 0      /* S e t t h i s t o N U L L w h e n u s i n g S R U T h e a d e r */

    e x t a d i t e r d a . f c e 5 p i n t l      = 0      /* D e p e c a t e d . S e t t o 0 */
    e x t a d i t e r d a a t t e m a t e D D L E = 0      /* D e p e c a t e d . S e t t o 0 */

    e s u l t = s y s _ e m - > i n t ( 3 2 , F M O D _ I N T _ N O R M A L , ( w i d t h ?
E R R _ E C K ( e s u l t ;

```

# SPURS

When using FMOD with SPURS, load the SPURS modules:

- /api/fmodex\_spurs.elf
- /api/fmodex\_spurs\_mpeg.elf

into 128 byte aligned memory and pass a pointer to this memory through the **spu\_mixer\_elfname\_or\_spursdata** and **spu\_streamer\_elfname\_or\_spursdata** members of the **FMOD\_PS3\_EXTRADRIVERDATA** structure.

You must also pass a pointer to your SPURS instance through **spurs** member of **FMOD\_PS3\_EXTRADRIVERDATA**.

```

F M O D _ P S 3 _ E X T R A D R I V E R D A T A e x t a d i t e r d a t a ;
C e l l s p u r s                                s p u r s ;
w i d t h                                         * f m o d _ x s p u r s _ b i t s
w i d t h                                         * f m o d _ x s p u r s _ m p e g _ b i t s

c e l l s y s m o d i f i e r d a t a ( C E L L _ S Y S M O D I F I E R _ S R U ) ;

s p u r s = ( C e l l s p u r s * ) m e m a l i g n ( C E L L _ S R U _ A L I G N , s i z e o f ( C e l l s p u r s ) ) ;

c e l l s p u r s i n i t i a l i z e ( s p u r s , 2 , 3 0 , p p _ t h r _ p r i o , f a l s e ) ;

    e x t a d i t e r d a s p _ m i x e r _ e l f n a m e _ o r s p u r s d a t a      =
_ b a y _ f m o d _ x s p u r s _ e l f s t a r t      /* P o i n t e r t o S R U S d a t a */
    e x t a d i t e r d a s p _ s t r e a m e r _ e l f n a m e _ o r s p u r s d a t a      =
_ b a y _ f m o d _ x s p u r s _ m p e g _ e l f s t a r t      /* P o i n t e r t o S R U S d a t a */

    m e m s e t (

    e x t a d i t e r d a s p _ m i x e r _ e l f n a m e _ o r s p u r s d a t a      = f m o d _ x s p u r s ;
    e x t a d i t e r d a s p _ s t r e a m e r _ e l f n a m e _ o r s p u r s d a t a      = f m o d _ x s p u r s _ m p e g _ b i t s

    e x t a d i t e r d a s p _ b a d _ f o m _ m e m o r y                                = 0      /* T H S W I L L B E I G N O R E D
*/

    e x t a d i t e r d a s p _ p r i o r i t y _ m i x e r                                = 16      /* D e f a u l t      T H S W I L L
B E I G N O R E D */
    e x t a d i t e r d a s p _ p r i o r i t y _ s t r e a m e r                        = 200     /* D e f a u l t      T H S W I L L
B E I G N O R E D */
    e x t a d i t e r d a s p _ p r i o r i t y _ a t t e m p t s                         = 200     /* D e f a u l t */

    e x t a d i t e r d a s p u r s = s p u r s ;

```

```

e x t a d i e r d a . f r e s p i n t l      = 0
e x t a d i e r d a a t e m a t e D D L E    = 0

e s u l t = s y s t e m - > i n t ( 3 2 , F M O D _ I N T _ D R M A L , ( w i d t h ?
E R R E C K ( e s u l t ;

```

For an example on using SPURS, please refer to the PlayStream example provided with the FMOD SDK.

## Loading Sounds into RSX Memory.

It is possible to load sounds into RSX memory using the [FMOD\\_LOADSECONDARYRAM](#) flag with [System::createSound](#).

In order to do this, you must pass a pointer to a pool of RSX memory through the **rsx\_pool** member in the **FMOD\_PS3\_EXTRADRIVERDATA** structure as well as the size of the RSX memory pool via **rsx\_pool\_size** member.

To get the usage of RSX memory pool, use the [System::getSoundRAM](#) function.

ie.

```

F M O D _ P S 3 _ E X T R A D R I V E R D A T A e x t a d i e r d a ;

w i d t h * s x _ m e m o y ;
i n t i d
i n t s x _ p o l _ s i z e = 3 2 * 1 0 2 4 * 1 0 2 4

/ *
    Some S X m e m o y
*/
g l b n d b f e r ( 1 , ?
g l b n d b f e r ( G L A R A Y _ B U F F E R , i d ) ;
g l b f e r d a ( G L A R A Y _ B U F F E R , s x _ p o l _ s i z e , N U L L , G L _ M A M I C _ D R A W ) ;
s x _ m e m o y = g m a p b f e r ( G L A R A Y _ B U F F E R , G L _ E A D W R E ) ;
g l b n d b f e r ( G L A R A Y _ B U F F E R , 0 ) ;

m e m s e t (

e x t a d i e r d a s p _ m i x r e l f a m e _ o r s p s d a      =
S Y S _ A P P _ D M E "/ f i n d x s p s e l f "
e x t a d i e r d a s p _ s t e a m e r e l f a m e _ o r s p s d a      =
S Y S _ A P P _ D M E "/ f i n d x s p _ m p g s e l f "

e x t a d i e r d a s p _ b a d f o m _ m e m o y                  = 0

e x t a d i e r d a s p _ p r o i t y _ m i x r                  = 1 6    / * d e f u l t */
e x t a d i e r d a s p _ p r o i t y _ a t s                   = 2 0 0  / * d e f u l t */
e x t a d i e r d a s p _ p r o i t y _ s t e a m e r           = 2 0 0  / * d e f u l t */

e x t a d i e r d a s p s = 0    / * S e t t h i s t o N U L L w h e n u s i n g S R U T h r e a d */

e x t a d i e r d a . s x _ p o l      = s x _ m e m o y ;          / * P o i n t e r t o S X m e m o y
p o l */
e x t a d i e r d a . s x _ p o l _ s i z e      = s x _ p o l _ s i z e ;          / * S i z e o f S X m e m o y p o l
*/

e x t a d i e r d a . f r e s p i n t l      = 0    / * d e p e c a t e d . S e t t o 0 */
e x t a d i e r d a a t e m a t e D D L E    = 0    / * d e p e c a t e d . S e t t o 0 */

e s u l t = s y s t e m - > i n t ( 3 2 , F M O D _ I N T _ D R M A L , ( w i d t h ?
E R R E C K ( e s u l t ;

```

```

/*
 *  ba dsou ndi nb  S Xmemo y
 */
esu lt= sys tē- 2 ea tSou nd(SYS_A PP_OME "/ dnm bō pm 3 ",
FMOD_ EA ECOM PRESSE SAM PE | FMO D_ DA SECO NIA E AM , 0, ?
E RE ECK (esu lt);

/*
 *  Ge t S XMemo y usage
 */
{
    i ntcu rē nt, ma x, b ā i

    sys tē- 2e tSou ndAm (?)
}

```

**Note:** There is a performance hit if using RSX memory for audio samples. There is about a 10% performance hit when playing mp3 samples, and about a 50% performance hit when playing PCM samples.

## FMOD\_ADVANCEDSETTINGS - maxPCMcodecs

The [FMOD\\_ADVANCEDSETTINGS](#) structure has a PS3 specific member, **maxPCMcodecs**. This sets the maximum number of PCM voices that can be played at once.

This includes streams of any format, as well as any sounds that are opened without the [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) flag, or sound formats not compatible with the [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) flag (compatible formats are mp2/mp3/adpcm). If not set, this defaults to 16.

Typically, all your sound effects would be in [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) and **maxPCMcodecs** would be set to the maximum number of streams you anticipate being played at once.

## Custom DSP Units

Custom DSP units are currently not supported on the PS3.

## AT3 Playback (NOT RECOMMENDED!)

**Note:** AT3 playback is now disabled and no longer supported.

AT3 playback makes use of Sony's own atrac library which is very inefficient. A new SPURS instance is created for each AT3 stream!! (this means 1 SPU per AT3 stream!). In fact, AT3 support has actually been abandoned by Sony. We highly discourage the use of this format and would recommend mp3/mp2/adpcm/pcm for the PS3.

## TRC: requirements for using the SPU selfs (fmodex\_spu.self / fmodex\_spu\_mpeg.self)

TRC 25.5 requires SPU programs that are not loaded from disk to be embedded, as raw .elf into the PPU program.

You can unsign the .self files back to .elf files using the "unself.exe" utility included in the Sony SDK, before embedding them into your program.

Set the **spu\_load\_from\_memory** member to 1 and point to the address of the embedded elfs using the **spu\_mixer\_elfname\_or\_spursdata** and **spu\_streamer\_elfname\_or\_spursdata** members of the **FMOD\_PS3\_EXTRADRIVERDATA** structure.

If you are just simply loading the SPU selfs from disk as per the above examples, you can just leave them as they are.

## TRC: requirements for using the SPURS elfs (fmodex\_spurs.elf / fmodex\_spurs\_mpeg.elf)

TRC 17.1.1 requires executable data on the Blu-Ray disc to be only in SELF/SPRX format.

If you are using SPURS you will need to embed the SPURS elf files into your application. The SPURS data can then be pointed to using the **spu\_mixer\_elfname\_or\_spursdata** and **spu\_streamer\_elfname\_or\_spursdata** members of the **FMOD\_PS3\_EXTRADRIVERDATA** structure.

The PlayStream example included in the FMOD SDK demonstrates how to load an embedded SPURS elf from the application.

For more information about this, please refer to the "Application Requirements" section of the Sony SDK documentation.

## TRC: Audio Formats Supported

The "Audio formats supported at boot" are as follows:

- 7.1ch LPCM
- 5.1ch LPCM, Downmix from 7.1ch LPCM
- 5.1ch Dolby Interactive Encoding
- 2ch LPCM, Downmix from 7.1ch LPCM

## Sounds only coming out of left/right speakers?

A common issue is that users incorrectly re-initialize the ps3 audio for movie players with settings that don't match what FMOD did. This causes problems with audio not coming out of all the speakers. You should not re-initialize the audio. The member **cell\_audio\_config** has been provided in **FMOD\_PS3\_EXTRADRIVERDATA** which gives information as to how FMOD has initialized PS3 audio. This information can be used for movie players etc.

## 7.1 surround speakers

On Playstation 3 in 7.1, the extra 2 speakers are not side left/side right, they are 'surround back left'/surround back right' which locate the speakers behind the listener instead of to the sides like on PC.

FMOD\_SPEAKER\_SBL/FMOD\_SPEAKER\_SBR are provided to make it clearer what speaker is being addressed on that platform.

## Running FMOD Examples.

All examples refer to files in "SYS\_APP\_HOME". "SYS\_APP\_HOME" should be set to the SDK media directory, **/examples/media**. In target manager for example, set the "Fileserving root directory" to **/examples/media**.

## What to do if audio cuts out or you get garbled sound.

There are two issues that usually cause this:

- If you have a thread that doesn't yield enough, a number of processes can die, including libAudio. This is usually observed during the loading of level data.

For details, please refer to <https://ps3.scedev.net/technotes/view/302>.

- If you are passing bad floats to FMOD. To check if this is the case, you should link to the logging version of FMOD (libfmodexL.a) and check to see if any functions are returning FMOD\_ERR\_INVALID\_FLOAT.

## Analog Audio Output Connections.

If using the analog audio output terminals of the DEH-R1040+ devkits, make sure "AV MULTI" is selected as the audio output in the bootup OSD. Otherwise the audio may sound "tinny".

Note: The analog audio output terminals are mis-labelled on some devkits, for details please refer to <https://ps3.scedev.net/technotes/view/160>.

# PLAYSTATION PORTABLE SPECIFIC ISSUES / FEATURES

## Installation.

You will need to use at least version 2.8.0 of the PSP SDK.

### Libraries.

Link this into your project. One of these files must be linked.

### GCC users.

- **/api/lib/fmodex.a** for general development with all possible features included. Software mixing features will incur higher CPU usage.
- **/api/lib/fmodexD.a** for the same library, but with debug logging which can help to determine any problems if they exist.
- **/api/lib/fmodex\_reduced.a** for general development with a smaller library size and features removed. See table below for which features are removed. Recommended for PSP, as it lowest memory usage and cpu usage.
- **/api/lib/fmodexD\_reduced.a** for the same library, but with debug logging which can help to determine any problems if they exist.

### SNC compiler users.

- **/api/lib/fmodex\_SNC.lib** for general development with all possible features included. Software mixing features will incur higher CPU usage.
- **/api/lib/fmodexD\_SNC.lib** for the same library, but with debug logging which can help to determine any problems if they exist.
- **/api/lib/fmodex\_reduced\_SNC.lib** for general development with a smaller library size and features removed. See table below for which features are removed. Recommended for PSP, as it lowest memory usage and cpu usage.
- **/api/lib/fmodexD\_reduced\_SNC.lib** for the same library, but with debug logging which can help to determine any problems if they exist.

Feature table.

Feature	fmodex.a	fmodex_reduced.a	Requires software mixing?
Streaming audio support	Y	Y	N
3D Sound	Y	Y	N
Virtual voices	Y	Y	N
FMOD Designer API support	Y	Y	N
Nonblocking sound open support	Y	Y	N
Hardware reverb	Y	Y	N
Geometry support / polygon occlusion	Y	N	N
Software mixing	Y	N	Y
Spectrum Analysis	Y	N	Y



Network streaming	N	N	n/a
Recording support	N	N	n/a
File format - FSB	Y	Y	N
File format - VAG	Y	Y	N
File format - AT3	Y	Y	N
File format - AIFF	N	N	Y
File format - DLS	Y	N	Y
File format - FLAC	N	N	Y
File format - IT (sequenced music format)	Y	N	Y
File format - MIDI (sequenced music format)	Y	N	Y
File format - MOD (sequenced music format)	Y	N	Y
File format - MP3	Y	Y	N
File format - Ogg Vorbis	N	N	Y
File format - M3U / PLS / ASX (Playlist format)	Y	N	Y
File format - RAW (format specified by user)	Y	N	Y
File format - S3M (sequenced music format)	Y	N	Y
File format - Tag formats - ID3V2, ASF, Ogg tags	N	N	N
File format - XM (sequenced music format)	Y	N	Y
File format - WAV	Y	N	Y
File format - User created	Y	Y	N
File format - ASF / WMA	N	N	n/a
File format - CDDA	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_PSP	Y	Y	N
Output mode - FMOD_OUTPUTTYPE_WAVWRITER	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_WAVWRITER_NRT	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_NOSOUND	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_NOSOUND_NRT	N	N	n/a
DSP Filter - Oscillator	Y	N	Y
DSP Effect - Lowpass	Y	N	Y
DSP Effect - Lowpass2	Y	N	Y
DSP Effect - Highpass	Y	N	Y
DSP Effect - Echo	Y	N	Y
DSP Effect - Flange	Y	N	Y
DSP Effect - Distortion	Y	N	Y
DSP Effect - Normalize	Y	N	Y
DSP Effect - Parameq	Y	N	Y
DSP Effect - Pitchshift	Y	N	Y
DSP Effect - Chorus	Y	N	Y
DSP Effect - Software reverb	Y	N	Y
DSP Effect - IT echo	Y	N	Y
DSP Effect - SFX reverb	N	N	n/a

In this table "**Requires software mixing?**" is specified to let the user know that the main CPU and RAM will be used to perform the feature which may not be desirable for the programmer.

Most things requiring the FMOD software mixer are removed in the reduced version of the library, to provide simple sound support.

The FMOD designer API can be used with the reduced library as well as long as all banks are marked as hardware in FMOD designer.

Note with a source code license you can easily turn features on and off to reduce code size or create different combinations of features to best suit your needs.

# Loading modules and linking libraries required for FMOD to operate.

FMOD requires the following sony libraries and modules to be used as well to get audio support. They are linked and loaded by the user.

- Link weak import stub library libsas\_weak.a into your project.
- Link weak import stub library libatrac3plus\_stub\_weak.a into your project.
- Link weak import stub library libmp3\_stub\_weak.a into your project.
- Load module SCE\_UTILITY\_AV\_MODULE\_SASCORE at runtime.

If you require AT3 playback support, load the following sony run-time modules:

- Load module SCE\_UTILITY\_AV\_MODULE\_LIBATRAC3PLUS
- Load module SCE\_UTILITY\_AV\_MODULE\_AVCODEC

If you require MP3 playback support, load the following sony run-time modules:

- Load module libmp3.prx
- Load module SCE\_UTILITY\_AV\_MODULE\_AVCODEC

## Loading the modules.

Load modules using the **sceUtilityLoadAvModule** / **sceUtilityUnloadAvModule** PSP SDK functions.

ie.

```
if (sceUtilityLoadAvModule(SCE_UTILITY_AV_MODULE_SASCORE)
```

bad the libmp3.prx module using the **sceKernelLoadModule** and **sceKernelStartModule** PSP SDK functions.

ie .

```
#define MODULE_NAME "bst0:cygwin/us r/bca l/ p p d k i t/module/libmp3.prx"
```

```
SceUID ModuleId  
SceIntr2 r
```

```
SceUID id = sceKernelLoadModule(MODULE_NAME, 0, NULL;  
if (id
```

**The E B f m a t - The e c o m m e n d e d  
f m a t f r s a m p l e s .**

Althugh .WAV and .W64 formats are supported, for badging speed it is highly recommended to use .SB files.

SB is hardware accelerated, WAV is not.

These are compressed batches of native PhysStation 0 raw sound data, arranged so when loaded, it is one read to bad the heads first, then the raw data (which is continuous), which is streamer friendly. This is the fastest way to bad sound data from UMD.

## Streaming ATAC music.

MO D supports the playback of a3 files using the SP hardware decoder (SP Media Engine).

By default, the maximum number of a3 decoder instances is 2, however it is possible to increase

this using the BPSX function `sceAtacBinit(a3_entry, i a3_plus_entry)`. The a3\_entry

parameter being the number of ATAC3s to be decoded simultaneously and the a3\_plus\_entry being the number

of ATAC3 plus' to be decoded simultaneously. The relationship between these two parameters is:

$a3\_entry + a3\_plus\_entry * 2 \leq 6$

Please refer to the BPSX documentation for more details.

## Streaming MB music.

MO D supports the playback of mp3 files using the SP hardware decoder (SP Media Engine). Only 44100 Hz mp3s are supported by the decoder.

The maximum number of mp3 decoder instances is 2, therefore it is not possible to play more than 2 mp3 streams at the same time.

## Anything to consider.

Note, that even though MO D supports streaming multiple streams from UMD to one, this is not recommended. On the PhysStation 0 raw seeks should be avoided at all times to preserve momentum of the umd head and therefore latency. This also goes for data streaming. Don't stream data and music at the same time if there is seeking involved. Continuous seeking will degrade latency because it has to mechanically move the seek head.

It may be preferable to play 'in memory' music such as sequenced formats like

MOD/S3M/.X or I Tor play music in a3 format. As ATAC can store a minute of audio per 128k of memory, then you would need 3mb of memory for 3 minutes of music.

# XBOX SPECIFIC ISSUES / FEATURES

## Installation

- Use **api/lib/fmodxbox.lib** for general development.
- Use **api/lib/fmodxboxD.lib** for the same library, but with logging which can help to determine any problems if they exist.

On FMOD XBox, you **must** call [FMOD::Memory\\_Initialize](#), and supply a pool of memory with a length.

for example.

```
# d e f i n e A U D I O _ M E M _ E N G T H ( 4 * 1 0 2 4 * 1 0 2 4  
  
c h a r * m e m = m a l l o c ( A U D I O _ M E M _ E N G T H ;  
  
FMOD::Memory\_Initialize ( m e m , A U D I O _ M E M _ E N G T H , N U L L , N U L L , N U L L ;
```

then call

[System::init](#).

The reasoning for this is for performance.

FMOD must be able to access sample data within its own memory block to avoid a slowdown issue in the DirectSound implementation on XBox.

**IDirectSoundBuffer8::SetBufferData** must create a table or an 'SGE list' every time this function is called, and it is slow.

Some games might call this function every time a sound was played, causing significant CPU degradation.

FMOD uses just 1 call to this function (at initialization time), and from then on uses the more efficient method of simply making Directsound XBox bufferdata pointers point to the whole FMOD memory block, and then from then on, it simply specifies an offset within that buffer for each hardware audio sound.

The memory provided must be enough to store all samples and extra system memory overhead for FMOD.

You can call [FMOD\\_Memory\\_GetStats](#) to determine what FMOD needs as a game runs.

You could run FMOD and supply it with an unrealistically high memory pool (say 8 megabytes), and then call [FMOD\\_Memory\\_GetStats](#) to determine the maximum amount of RAM fmod needs to store sounds and for FMOD system overhead.

## The 8mb Memory Limitation.

Currently for hardware sound effects, there is an 8mb limit for sound effects.  
This is due to the XBox DirectSound architecture.

From the XDK Documentation : *"DirectSound buffers are managed in a scatter gather entry (SGE) list. There is a maximum of 2,047 SGEs, which each point to a 4-KB page. This means that a maximum of 8 MB are available for allocating or playing DirectSound buffers simultaneously"*

Future versions may have multiple 8mb pools if it is required by developers but generally this is more than enough for audio.

Remember that data stored within this memory can be XADPCM format which is about 3.5:1 compression.

## WMA support.

WMA is supported but generally as a streaming format. The benefits of WMA vs Ogg vorbis for example are negligible, so you could use whatever audio format you like for music streaming.

## Formats not supported.

Currently all advertised FMOD formats are supported. This may change as some formats are not used generally (ie .FLAC) and just take up code space that they don't need to.

**Note!** With a source code license you can remove and add whatever formats you like, which will reduce the size of the library significantly. For example you can easily remove all formats except for XMA if so desired.

## 5.1 support and speaker settings.

Note that the Xbox dashboard is the only place the speaker settings are selected. This is done by the user and should not be changed or forced by the code in any way, as it will go against the user's selection.

FMOD will automatically use the correct speaker setting that was selected in the dashboard.

# XBOX 360 SPECIFIC ISSUES / FEATURES

## Installation

- Use **api/lib/fmodxbox360.lib** for general development.
- Use **api/lib/fmodxbox360D.lib** for the same library, but with debug logging which can help to determine any problems if they exist.

FMOD also uses some XDK libraries. You must link with the following.

- **xmp.lib** for XMPGetStatus to determine if the dashboard is playing its own music or not.
- **xaudio.lib** for FMOD sound output support.

Besides this there are no other requirements. It is optional for you to give fmod a block of memory to work within if you like (ie using [FMOD::Memory\\_Initialize](#)), otherwise you can let FMOD simply use the default memory allocators provided by the Xbox 360 operating system.

## Memory

This is an important subject, as performance can be bad or even worse, FMOD will not function correctly.

XMA support needs buffers allocated with XPhysicalAlloc. malloc and free will not allow XMA to function correctly.

By default FMOD uses XPhysicalAlloc to allocate memory, but the page size for XPhysicalAlloc is 4096 bytes. As FMOD can do a lot of smaller allocations, this is grossly inefficient and wastes memory.

The solution is to use your own memory manager, or let FMOD manage the memory with [FMOD::Memory\\_Initialize](#). If you have an efficient, low page size memory manager, that can use memory allocated with XPhysicalAlloc, use the memory callback feature of [FMOD::Memory\\_Initialize](#).

If not, the easier solution is to allocate just 1 large block of memory with XPhysicalAlloc, then simply give it to FMOD via [FMOD::Memory\\_Initialize](#) (leave callbacks set to NULL). With this you can simply pass the buffer and size, and FMOD will manage the memory internally and not allocate outside of this.

FMOD's memory manager has a 32byte page size. If your memory manager has pages bigger than this, it would be more efficient to use fmod's memory manager.

## Multiple CPUs

As the Xbox 360 has multiple cpus, you can specify which cpu and which hardware thread FMOD Ex's threads can operate on.

Before doing this though, note that FMOD has already selected **Thread 4** (CORE2, HW Thread 0) to process its software mixer thread, stream thread, [FMOD\\_NONBLOCKING](#) loader thread, and file thread.

We chose this CPU and thread as this is the same CPU and thread that XAudio runs on by default. You will not necessarily have to change it, because it won't affect the main game code which is assumed to be running on **Thread 0**.

If you want to change FMOD's core and hardware thread assignments, just use the structure found in **fmodxbox360.h**, which is in api/inc.

This structure is then passed in as the 'extradrivervdata' parameter of [System::init](#).

## Dashboard music technical requirement.

Xbox 360 TCR states that you must allow the user to select their own music from the dashboard, which should then in turn make the built in game music go quiet or pause.

To do this is simple with FMOD Ex. The mechanism used is for the user to create a special channel group (see [System::createChannelGroup](#)) with the name "**music**".

When you play a music track, set the channel group for the channel to this music group, and FMOD will automatically pause it if the user selects a song from the dashboard to play.

If using the "FMOD Designer" sound designer tool, put all music events under the "**master/music**" category (or one of its sub-categories).

## XMA support.

To load a .XMA file as a static compressed sample, the [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) or [FMOD\\_CREATECOMPRESSEDMEMSAMPLE](#) must be used when loading the file.

## Formats not supported.

Currently only .FLAC has been removed, as it is generally an uncommon format to be needed on a console such as the Xbox 360, and its inclusion adds too much to the library size.

**Note!** With a source code license you can remove and add whatever formats you like, which will reduce the size of the library significantly. For example you can easily remove all formats except for XMA if so desired.

## 5.1 support and speaker settings.

Note that the Xbox dashboard is the only place the speaker settings are selected. This is done by the user and should not be changed or forced by the code in any way, as it will go against the user's selection.

FMOD will automatically use the correct speaker setting that was selected in the dashboard.

The programmer is to assume everything is in 5.1 internally, and not be concerned about the end users' setup.

# GAMECUBE SPECIFIC ISSUES / FEATURES

## Installation

Linking FMOD Ex to your application.

- **/api/lib/fmodgc.lib** - Link to this file if you are using **SN Systems** compiler.
- **/api/lib/fmodgcD.lib** - This is the debug version of fmodgc.lib and outputs a log of FMOD's progress and any error messages (in plain english) to the TTY.
- **/api/lib/fmodgc\_cw.a** - Link to this file if you are using the **Metrowerks Codewarrior** compiler.
- **/api/lib/fmodgc\_cwD.a** - This is the debug version of fmodgc\_cw.lib and outputs a log of FMOD's progress and any error messages (in plain english) to the TTY.

## Running the examples.

Simply load their .dsp files into Dev Studio and hit F7.

**NOTE:** You will need to copy all files in the media directory to  
\$DVDROOT/fmod. For example:

```
copy media\*.* c:\DolphinSDK1.0\dvddata\fmod
```

## Formats not supported.

WMA is not support on FMOD Ex for Gamecube. This is because FMOD uses a windows codec to be able to decode WMA. This codec is proprietary and owned by Microsoft and is not cross platform.

FLAC is not supported on FMOD Ex for GameCube. The FLAC codec is rarely used and just takes up unecessary code space. If it is need then contact us at [support@fmod.org](mailto:support@fmod.org).



# Wii SPECIFIC ISSUES / FEATURES

## Installation.

### Libraries.

Link this into your project. One of these files must be linked.

- **/api/lib/fmodwii.a** for general development with all possible features included.
- **/api/lib/fmodwiiL.a** for the same library, but with debug logging which can help to determine any problems if they exist.
- **/api/lib/fmodwii\_reduced.a** for general development with a smaller library size and features removed. See table below for which features are removed.
- **/api/lib/fmodwiiL\_reduced.a** for the same library, but with debug logging which can help to determine any problems if they exist.

The following libraries should also be linked into your project:

- base.a
- os.a
- exi.a
- si.a
- db.a
- vi.a
- gx.a
- dvd.a
- ai.a
- ax.a
- axfx.a
- mix.a
- dsp.a
- ipc.a
- fs.a
- pad.a
- wpad.a
- wenc.a
- wud.a
- euart.a
- sc.a
- nand.a

Feature table.

Feature	fmodwii.a	fmodwii_reduced.a	Requires software mixing?
Streaming audio support	Y	Y	N

3D Sound	Y	Y	N
Virtual voices	Y	Y	N
FMOD Designer API support	Y	Y	N
Nonblocking sound open support	Y	Y	N
Hardware reverb	Y	Y	N
Geometry support / polygon occlusion	Y	N	N
Software mixing	Y	N	Y
Spectrum Analysis	Y	N	Y
Network streaming	N	N	n/a
Recording support	N	N	n/a
File format - FSB	Y	Y	N
File format - DSP	Y	Y	N
File format - AIFF	Y	N	Y
File format - DLS	Y	N	Y
File format - FLAC	N	N	Y
File format - IT (sequenced music format)	Y	N	Y
File format - MIDI (sequenced music format)	Y	N	Y
File format - MOD (sequenced music format)	Y	N	Y
File format - MP2 / MP3	Y	N	Y
File format - Ogg Vorbis	Y	N	Y
File format - M3U / PLS / ASX (Playlist format)	Y	N	Y
File format - RAW (format specified by user)	Y	N	Y
File format - S3M (sequenced music format)	Y	N	Y
File format - Tag formats - ID3V2, ASF, Ogg tags	Y	N	Y
File format - XM (sequenced music format)	Y	N	Y
File format - WAV	Y	N	Y
File format - User created	Y	Y	N
File format - ASF / WMA	N	N	n/a
File format - CDDA	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_WII	Y	Y	N
Output mode - FMOD_OUTPUTTYPE_WAVWRITER	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_WAVWRITER_NRT	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_NOSOUND	N	N	n/a
Output mode - FMOD_OUTPUTTYPE_NOSOUND_NRT	N	N	n/a
DSP Filter - Oscillator	Y	N	Y
DSP Effect - Lowpass	Y	N	Y
DSP Effect - Lowpass2	Y	N	Y
DSP Effect - Highpass	Y	N	Y
DSP Effect - Echo	Y	N	Y
DSP Effect - Flange	Y	N	Y
DSP Effect - Distortion	Y	N	Y
DSP Effect - Normalize	Y	N	Y
DSP Effect - Parameq	Y	N	Y
DSP Effect - Pitchshift	Y	N	Y
DSP Effect - Chorus	Y	N	Y
DSP Effect - Software reverb	Y	N	Y
DSP Effect - IT echo	Y	N	Y

In this table "**Requires software mixing?**" is specified to let the user know that the main CPU and RAM will be used to perform the feature which may not be desirable for the programmer.

Most things requiring the FMOD software mixer are removed in the reduced version of the library, to provide simple sound support.

The FMOD designer API can be used with the reduced library as well as long as all banks are marked as hardware in FMOD designer.

Note with a source code license you can easily turn features on and off to reduce code size or create different combinations of features to best suit your needs.

## DVD Read Priority

It is possible to set the DVD read priority using the **dvdreadpriority** member in the **FMOD\_WII\_INFO** structure. A pointer to this structure is passed to [System::init](#) as the **extradrivedata** parameter. The setting can be in the range 0 - 3 with 0 being the highest and 3 the lowest. The default setting is 1.

## Wii Controller Speaker Support

FMOD supports the playing of sounds out of the Wii controller speaker using the Wii specific API functions found in **fmodwii.h**.

The **FMOD\_Wii\_Controller\_Command** function can be used to turn the Wii controller speaker on and off using the **FMOD\_WII\_CONTROLLER\_CMD\_SPEAKERON** and **FMOD\_WII\_CONTROLLER\_CMD\_SPEAKEROFF** commands.

Use the **FMOD\_Channel\_SetControllerSpeaker** function to set a channel to play out of the controller speaker. The last argument of the function, of type **FMOD\_WII\_CONTROLLER**, specifies which controller the channel is to be played out of. These can be or'd together to play a single channel out of multiple controllers. You can set this argument to **FMOD\_WII\_CONTROLLER\_NONE** on a channel currently playing out of a controller speaker to just play out of regular speakers again.

Example of usage:

```
/*
 * Turn on the controller speaker. This should be done if you want to connect
 * call back, or
 * when you know the controller is connected.
 */
FMOD_Wii_Controller_Command(hn, FMOD_WII_CONTROLLER_CMD_SPEAKERON);

.
.
.

/*
 * Play a sound with the channel played
 */
result = system->playSound(FMOD_CHANNEL_FREE, sound, true, ?
ENGINE_CHECK(result);

/*
 * Set the channel to play out of FMOD_WII_CONTROLLER_0
 */
result = FMOD_Channel_SetControllerSpeaker((FMOD_CHANNEL *)channel,
FMOD_WII_CONTROLLER_0;
ENGINE_CHECK(result);
```

```

/*
    Unpause channel
*/
esult = channel->SetPaused(false);
ERR_CHECK(esult);

```

The **controllerspeaker** example demonstrates the usage of the Wii controller speaker.

It is possible to have FMOD mute/unmute the controller speaker depending on whether a sound is playing. This may help to reduce the battery life of the controller. This feature can be turned on by setting the **muteremotespeakerifnosound** member in the **FMOD\_WII\_INFO** structure to 1.

## Using MEM2 for FMOD.

By default, all memory allocations in FMOD will use OSAlloc which will allocate from the current heap. If you wish to use MEM2, you may allocate a block of memory in MEM2 and pass it to FMOD to use.  
ie.

```

void          aMem2b;
void          aMem2h;
MEM_Handle    xpMap

/*
    Allocate a block of memory for FMOD to use in MEM2 (optional)
*/
aMem2b = OSGetMEM2Addr();
aMem2h = OSGetMEM2Addr();
xpMap = MEMCHandleExpMap(aMem2b, (u32)aMem2h - (u32)aMem2b);

memBlock = MEMAllocFromExpMap(xpMap, 16 * 1024 * 1024);

FMOD_Memory_Init(&memBlock, 16*1024*1024, NULL, NULL, NULL);

```

## Handling disk eject.

When playing a stream and the disk is ejected, it will keep re-trying the read automatically until the disk is re-inserted. The **starving** parameter in function [Sound::getOpenState](#) will let you know when this is the case and you can mute the stream accordingly to prevent stutters.

If are calling functions that read from disk in the main thread, or using the [FMOD\\_NONBLOCKING](#) flag, you will get the [FMOD\\_ERR\\_FILE\\_DISKEJECTED](#) error returned.

# API Reference

[FMOD Ex API Reference](#)

[FMOD Designer API Reference](#)

[FMOD Designer Network API Reference](#)

# C++ Reference

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# Interfaces

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[SoundGroup](#)

[DSP](#)

[Geometry](#)

[Reverb](#)

# System Interface

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[System::getDriverInfo](#)  
[System::getGeometrySettings](#)  
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[System::getMasterChannelGroup](#)  
[System::getMasterSoundGroup](#)  
[System::getNetworkProxy](#)  
[System::getNetworkTimeout](#)  
[System::getNumCDROMDrives](#)  
[System::getNumDrivers](#)  
[System::getNumPlugins](#)  
[System::getOutput](#)  
[System::getOutputByPlugin](#)  
[System::getOutputHandle](#)  
[System::getPluginInfo](#)  
[System::getRecordDriver](#)  
[System::getRecordDriverCaps](#)  
[System::getRecordDriverInfo](#)  
[System::getRecordNumDrivers](#)  
[System::getRecordPosition](#)  
[System::getReverbAmbientProperties](#)  
[System::getReverbProperties](#)



[System::getSoftwareChannels](#)  
[System::getSoftwareFormat](#)  
[System::getSoundRAM](#)  
[System::getSpeakerMode](#)  
[System::getSpectrum](#)  
[System::getStreamBufferSize](#)  
[System::getUserData](#)  
[System::getVersion](#)  
[System::getWaveData](#)  
[System::init](#)  
[System::isRecording](#)  
[System::loadGeometry](#)  
[System::loadPlugin](#)  
[System::lockDSP](#)  
[System::playDSP](#)  
[System::playSound](#)  
[System::recordStart](#)  
[System::recordStop](#)  
[System::release](#)  
[System::set3DListenerAttributes](#)  
[System::set3DNumListeners](#)  
[System::set3DRolloffCallback](#)  
[System::set3DSettings](#)  
[System::set3DSpeakerPosition](#)  
[System::setAdvancedSettings](#)  
[System::setCallback](#)  
[System::setDSPBufferSize](#)  
[System::setDriver](#)  
[System::setFileSystem](#)  
[System::setGeometrySettings](#)  
[System::setHardwareChannels](#)  
[System::setNetworkProxy](#)  
[System::setNetworkTimeout](#)  
[System::setOutput](#)  
[System::setOutputByPlugin](#)  
[System::setPluginPath](#)  
[System::setRecordDriver](#)  
[System::setReverbAmbientProperties](#)  
[System::setReverbProperties](#)  
[System::setSoftwareChannels](#)  
[System::setSoftwareFormat](#)  
[System::setSpeakerMode](#)  
[System::setStreamBufferSize](#)  
[System::setUserData](#)  
[System::unloadPlugin](#)  
[System::unlockDSP](#)  
[System::update](#)

# System::addDSP

This function adds a pre-created DSP unit or effect to the head of the System DSP chain.

## Syntax

```
FMOD_RESULT System::addDSP(
    FMOD_DSP * dsp
);
```

## Parameters

*dsp*

A pointer to a pre-created DSP unit to be inserted at the head of the System DSP chain.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is a wrapper function to insert a DSP unit at the top of the System DSP chain.

It disconnects the head unit from its input, then inserts the unit at the head and reconnects the previously disconnected input back as an input to the new unit.

It is effectively the following code.

```
int units;
system->getDSPHead(
    &head & getMmiUnits()
if (units > 1)
{
    return FMOD_ERR_DSP_INVALIDCONNECTION;
}
&head->disconnectFrom(next);
&head->addUnit(&p);
&p->addUnit(next);
&p->setActive(true);
```

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::getDSPHead](#)
- [\\_System::createDSP](#)
- [\\_System::createDSPByType](#)
- [\\_System::createDSPByIndex](#)
- [\\_Channel::addDSP](#)
- [\\_ChannelGroup::addDSP](#)
- [\\_DSP::remove](#)

# System::attachFileSystem

Function to allow a user to 'piggyback' on FMOD's file reading routines. This allows users to capture data as FMOD reads it, which may be useful for ripping the raw data that FMOD reads for hard to support sources (for example internet streams or cdda streams).?

## Syntax

```
FMOD_RESULT System::attachFileSystem (
    FMOD_FILE_OPEN_CALLBACK useOpen,
    FMOD_FILE_CLOSE_CALLBACK useClose,
    FMOD_FILE_READ_CALLBACK useRead,
    FMOD_FILE_SEEK_CALLBACK useSeek
);
```

## Parameters

*useOpen*

Pointer to an open callback which is called after a file is opened by FMOD.

*useClose*

Pointer to a close callback which is called after a file is closed by FMOD.

*useRead*

Pointer to a read callback which is called after a file is read by FMOD.

*useSeek*

Pointer to a seek callback which is called after a file is seeked into by FMOD.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setFileSystem](#)
- [FMOD\\_FILE\\_OPEN\\_CALLBACK](#)
- [FMOD\\_FILE\\_CLOSE\\_CALLBACK](#)

- [FMOD\\_FILE\\_READCALLBACK](#)
- [FMOD\\_FILE\\_SEEKCALLBACK](#)

Version 4.12.03 Built on Feb 18, 2008

# System::close

Closes the system object without freeing the object's memory, so the system handle will still be valid.  
?Closing the output renders objects created with this system object invalid. Make sure any sounds, channelgroups, geometry and dsp objects are released before closing the system object.  
?

## Syntax

```
FMOD_RESULT Sys::close();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).  
If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::init](#)
- [System::release](#)

# System::createChannelGroup

Creates a channel group object. These objects can be used to assign channels to for group channel settings, such as volume.

?Channel groups are also used for sub-mixing. Any channels that are assigned to a channel group get submixed into that channel group's DSP.?

## Syntax

```
FMOD_RESULT Sys::createChannelGroup(
    const char * name,
    FMOD_CHANNELGROUP ** channelgroup
);
```

## Parameters

*name*

Label to give to the channel group for identification purposes. Optional (can be null).

*channelgroup*

Address of a variable to receive a newly created FMOD::ChannelGroup object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

See the channel group class definition for the types of operations that can be performed on 'groups' of channels. The channel group can for example be used to have 2 separate groups of master volume, instead of one global master volume.

A channel group can be used for sub-mixing, ie so that a set of channels can be mixed into a channel group, then can have effects applied to it without affecting other channels.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getMasterChannelGroup](#)
- [Channel::setChannelGroup](#)

- [\\_ChannelGroup::release](#)

Version 4.12.03 Built on Feb 18, 2008



# System::createCodec

Creates an in memory file format codec to be used by FMOD by passing in a codec description structure.  
Once this is created, FMOD will use it to open user defined file formats.

## Syntax

```
FMOD_RESULT System::createCodec (
    FMOD_CODEC_DESCRIPTION * description
);
```

## Parameters

*description*

Address of a [FMOD\\_CODEC\\_DESCRIPTION](#) structure, containing information about the codec.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_CODEC\\_DESCRIPTION](#)

# System::createDSP

Creates a user defined DSP unit object to be inserted into a DSP network, for the purposes of sound filtering or sound generation.?

## Syntax

```
FMOD_RESULT System::createDSP(
    FMOD_DSP_DESCRIPTION *description,
    FMOD_DSP **dsp
);
```

## Parameters

*description*

Address of an [FMOD\\_DSP\\_DESCRIPTION](#) structure containing information about the unit to be created.

*dsp*

Address of a variable to receive a newly created FMOD::DSP object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A DSP unit can generate or filter incoming data.

The data is created or filtered through use of the read callback that is defined by the user.

See the definition for the [FMOD\\_DSP\\_DESCRIPTION](#) structure to find out what each member means.

To be active, a unit must be inserted into the FMOD DSP network to be heard. Use functions such as [System::addDSP](#), [Channel::addDSP](#) or [DSP::addInput](#) to do this.

For more information and a detailed description (with diagrams) see the tutorial on the DSP system in the documentation.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_DSP\\_DESCRIPTION](#)
- [System::createDSPByType](#)

- [\\_System::createDSPByIndex](#)
- [\\_System::addDSP](#)
- [\\_Channel::addDSP](#)
- [\\_DSP::addInput](#)
- [\\_DSP::setActive](#)

Version 4.12.03 Built on Feb 18, 2008

# System::createDSPByIndex

Creates a DSP unit object which is either built in or loaded as a plugin, to be inserted into a DSP network, for the purposes of sound filtering or sound generation.

?This function creates a DSP unit that can be enumerated by using [System::getNumPlugins](#) and [System::getPluginInfo](#).?

## Syntax

```
FMOD_RESULT System::createDSPByIndex(
    int index,
    FMOD_DSP** dsp
);
```

## Parameters

*index*

The index into the list of enumerable DSP plugins to create.

*dsp*

Address of a variable to receive a newly created FMOD::DSP object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A DSP unit can generate or filter incoming data.

To be active, a unit must be inserted into the FMOD DSP network to be heard. Use functions such as [System::addDSP](#), [Channel::addDSP](#) or [DSP::addInput](#) to do this.

For more information and a detailed description (with diagrams) see the tutorial on the DSP system in the documentation.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getNumPlugins](#)
- [System::getPluginInfo](#)

- [\\_System::createDSPByType](#)
- [\\_System::createDSP](#)
- [\\_System::addDSP](#)
- [\\_Channel::addDSP](#)
- [\\_DSP::addInput](#)
- [\\_DSP::setActive](#)

Version 4.12.03 Built on Feb 18, 2008

# System::createDSPByType

Creates an FMOD defined built in DSP unit object to be inserted into a DSP network, for the purposes of sound filtering or sound generation.

?This function is used to create special effects that come built into FMOD.?

## Syntax

```
FMOD_RESULT System::createDSPByType (
    FMOD_DSP_TYPE type,
    FMOD::DSP** dsp
);
```

## Parameters

*type*

A pre-defined DSP effect or sound generator described by a [FMOD\\_DSP\\_TYPE](#).

*dsp*

Address of a variable to receive a newly created FMOD::DSP object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A DSP unit can generate or filter incoming data.

To be active, a unit must be inserted into the FMOD DSP network to be heard. Use functions such as [System::addDSP](#), [Channel::addDSP](#), [ChannelGroup::addDSP](#) or [DSP::addInput](#) to do this.

For more information and a detailed description (with diagrams) see the tutorial on the DSP system in the documentation.

**Note!** Winamp DSP and VST plugins will only return the first plugin of this type that was loaded!

To access all VST or Winamp DSP plugins the [System::createDSPByIndex](#) function! Use the index returned by [System::loadPlugin](#) if you don't want to enumerate them all.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_DSP\\_TYPE](#)
- [System::createDSP](#)
- [System::createDSPByIndex](#)
- [System::addDSP](#)
- [System::loadPlugin](#)
- [Channel::addDSP](#)
- [ChannelGroup::addDSP](#)
- [DSP::addInput](#)
- [DSP::setActive](#)

# System::createGeometry

Geometry creation function. This function will create a base geometry object which can then have polygons added to it.

## Syntax

```
FMOD_RESULT System::createGeometry (
    int maxpolygons,
    int maxvertices,
    FMOD::Geometry ** geometry
);
```

## Parameters

*maxpolygons*

Maximum number of polygons within this object.

*maxvertices*

Maximum number of vertices within this object.

*geometry*

Address of a variable to receive a newly created FMOD::Geometry object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Polygons can be added to a geometry object using [Geometry::AddPolygon](#).

A geometry object stores its list of polygons in a structure optimized for quick line intersection testing and efficient insertion and updating.

The structure works best with regularly shaped polygons with minimal overlap.

Many overlapping polygons, or clusters of long thin polygons may not be handled efficiently.

Axis aligned polygons are handled most efficiently.

The same type of structure is used to optimize line intersection testing with multiple geometry objects.

It is important to set the value of maxworldsize to an appropriate value using [System::setGeometrySettings](#).

Objects or polygons outside the range of maxworldsize will not be handled efficiently.

Conversely, if maxworldsize is excessively large, the structure may lose precision and efficiency may drop.



## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::setGeometrySettings](#)
- [\\_System::loadGeometry](#)
- [\\_Geometry::AddPolygon](#)

Version 4.12.03 Built on Feb 18, 2008

# System::createReverb

Creates a 'virtual reverb' object. This object reacts to 3d location and morphs the reverb environment based on how close it is to the reverb object's center.

?Multiple reverb objects can be created to achieve a multi-reverb environment.?

## Syntax

```
FMOD_RESULT System::createReverb(
    FMOD::Reverb ** pReverb
);
```

## Parameters

*reverb*

Address of a pointer to a Reverb object to receive the newly created virtual reverb object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 3D reverb object is a sphere having 3D attributes (position, minimum distance, maximum distance) and reverb properties.

The properties and 3D attributes of all reverb objects collectively determine, along with the listener's position, the settings of and input gains into a single 3D reverb DSP.

Please note that this only applies to software channels. When the listener is within the sphere of effect of one or more 3d reverbs, the listener's 3D reverb properties are a weighted combination of such 3d reverbs. When the listener is outside all of the reverbs, the 3D reverb setting is set to the default ambient reverb setting.

Use [System::setReverbAmbientProperties](#) to set a 'background' default reverb environment. This is a reverb that will be morphed to if the listener is not within any virtual reverb zones.

By default the ambient reverb is set to 'off'.

Creating multiple reverb objects does not impact performance. These are 'virtual reverbs'. There will still be only 1 physical reverb DSP running that just morphs between the different virtual reverbs.

[System::setReverbProperties](#) can still be used in conjunction with the 3d based virtual reverb system. This allows 2d sounds to have reverb. If this call is used at the same time virtual reverb objects are active, 2 physical reverb dsp's will be used, incurring a small memory and cpu hit.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Reverb::release](#)
- [System::setReverbAmbientProperties](#)
- [System::getReverbAmbientProperties](#)
- [System::setReverbProperties](#)
- [System::getReverbProperties](#)

# System::createSound

Loads a sound into memory, or opens it for streaming.?

## Syntax

```
FMOD_RESULT Sys::createSound(
    const char * name_or_data,
    FMOD_MODE mode,
    FMOD_CREATESOUNDEXINFO * exinfo,
    FMOD::Sound ** sound
);
```

## Parameters

*name\_or\_data*

Name of the file or URL to open, or pointer to memory block if [FMOD\\_OPENMEMORY/](#)[FMOD\\_OPENMEMORY\\_POINT](#) is used. For CD playback the name should be a drive letter with a colon, example "D:" (windows only).

*mode*

Behaviour modifier for opening the sound. See [FMOD\\_MODE](#). Also see remarks for more.

*exinfo*

Pointer to a [FMOD\\_CREATESOUNDEXINFO](#) which lets the user provide extended information while playing the sound. Optional. Specify 0 or NULL to ignore.

*sound*

Address of a variable to receive a newly created FMOD::Sound object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

**Important!** By default ([FMOD\\_CREATESAMPLE](#)) FMOD will try to load and decompress the whole sound into memory! Use [FMOD\\_CREATESTREAM](#) to open it as a stream and have it play back in realtime! [FMOD\\_CREATECOMPRESSED](#) can also be used for certain formats.

- To open a file or URL as a stream, so that it decompresses / reads at runtime, instead of loading / decompressing into memory all at the time of this call, use the [FMOD\\_CREATESTREAM](#) flag. This is like a 'stream' in FMOD 3.
- To open a file or URL as a compressed sound effect that is not streamed and is not decompressed into memory

at load time, use [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#). This is supported with MPEG (mp2/mp3), ADPCM (wav on all platforms and vag on ps2/psp) and XMA files only. This is useful for those who want realtime compressed sound effects, but not the overhead of disk access.

- To open a CD drive, use the drive as the name, for example on the windows platform, use "D:"
- To open a sound as 2D, so that it is not affected by 3D processing, use the [FMOD\\_2D](#) flag. 3D sound commands will be ignored on these types of sounds.
- To open a sound as 3D, so that it is treated as a 3D sound, use the [FMOD\\_3D](#) flag. Calls to [Channel::setPan](#) will be ignored on these types of sounds.
- To use FMOD software mixing buffers, use the [FMOD\\_SOFTWARE](#) flag. This gives certain benefits, such as DSP processing, spectrum analysis, loop points, 5.1 mix levels, 2d/3d morphing, and more.
- To use the soundcard's hardware to play the sound, use the [FMOD\\_HARDWARE](#) flag. This gives certain benefits such as EAX reverb, dolby digital output on some devices, and better 3d sound virtualization using headphones.

Note that [FMOD\\_OPENRAW](#), [FMOD\\_OPENMEMORY](#), [FMOD\\_OPENMEMORY\\_POINT](#) and [FMOD\\_OPENUSER](#) will not work here without the exinfo structure present, as more information is needed.

Use [FMOD\\_NONBLOCKING](#) to have the sound open or load in the background. You can use [Sound::getOpenState](#) to determine if it has finished loading / opening or not. While it is loading (not ready), sound functions are not accessible for that sound.

To account for slow devices or computers that might cause buffer underrun (skipping/stuttering/repeating blocks of audio), use [System::setStreamBufferSize](#).

To play WMA files on Windows, the user must have the latest Windows media player codecs installed (Windows Media Player 9). The user can download this as an installer (wmfdist.exe) from [www.fmod.org](http://www.fmod.org) download page if they desire or you may wish to redistribute it with your application (this is allowed). This installer does NOT install windows media player, just the necessary WMA codecs needed.

**PlayStation 2 Note:** You can pre-pend "host0:" or "cdrom0:" if you like. FMOD will automatically add "host0:" to the filename if it is not found.

Specifying [FMOD\\_OPENMEMORY\\_POINT](#) will POINT to your memory rather allocating its own sound buffers and duplicating it internally

**This means you cannot free the memory while FMOD is using it, until after Sound::release is called.** With [FMOD\\_OPENMEMORY\\_POINT](#), for PCM formats, only WAV, FSB and RAW are supported. For compressed formats, only those formats supported by [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) are supported.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MODE](#)
- [FMOD\\_CREATEINDEXINFO](#)
- [Sound::getOpenState](#)
- [System::setStreamBufferSize](#)
- [Channel::setPan](#)



# System::createSoundGroup

Creates a sound group, which can store handles to multiple Sound pointers.?

## Syntax

```
FMOD_RESULT System::createSoundGroup(
    const char * name,
    FMOD_SOUNDGROUP ** soundgroup
);
```

## Parameters

*name*

Name of sound group.

*soundgroup*

Address of a variable to receive a pointer to a sound group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Once a SoundGroup is created, [Sound::setSoundGroup](#) is used to put a sound in a SoundGroup.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::release](#)
- [Sound::setSoundGroup](#)

# System::createStream

Opens a sound for streaming. This function is a helper function that is the same as [System::createSound](#) but has the [FMOD\\_CREATESTREAM](#) flag added internally.?

## Syntax

```
FMOD_RESULT System::createStream (
    const char * name_or_data ,
    FMOD_MODE mode ,
    FMOD_CREATESOUNDEXINFO * exinfo ,
    FMOD_SOUND ** sound
);
```

## Parameters

*name\_or\_data*

Name of the file or URL to open. For CD playback this may be a drive letter with a colon, example "D:".

*mode*

Behaviour modifier for opening the sound. See [FMOD\\_MODE](#). Also see remarks for more.

*exinfo*

Pointer to a [FMOD\\_CREATESOUNDEXINFO](#) which lets the user provide extended information while playing the sound. Optional. Specify 0 or NULL to ignore.

*sound*

Address of a variable to receive a newly created FMOD::Sound object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note that a stream only has 1 decode buffer and file handle, and therefore can only be played once. It cannot play multiple times at once because it cannot share a stream buffer if the stream is playing at different positions. Open multiple streams to have them play concurrently.

- To open a file or URL as a stream, so that it decompresses / reads at runtime, instead of loading / decompressing into memory all at the time of this call, use the [FMOD\\_CREATESTREAM](#) flag. This is like a 'stream' in FMOD 3.
- To open a file or URL as a compressed sound effect that is not streamed and is not decompressed into memory at load time, use [FMOD\\_CREATECOMPRESSED](#). This is supported with MPEG (mp2/mp3),



ADPCM (wav on all platforms and vag on ps2/psp) and XMA files only. This is useful for those who want realtime compressed soundeffects, but not the overhead of disk access.

- To open a CD drive, use the drive as the name, for example on the windows platform, use "D:"
- To open a sound as 2D, so that it is not affected by 3D processing, use the [FMOD\\_2D](#) flag. 3D sound commands will be ignored on these types of sounds.
- To open a sound as 3D, so that it is treated as a 3D sound, use the [FMOD\\_3D](#) flag. Calls to [Channel::setPan](#) will be ignored on these types of sounds.
- To use FMOD software mixing buffers, use the [FMOD\\_SOFTWARE](#) flag. This gives certain benefits, such as DSP processing, spectrum analysis, loop points, 5.1 mix levels, 2d/3d morphing, and more.
- To use the soundcard's hardware to play the sound, use the [FMOD\\_HARDWARE](#) flag. This gives certain benefits such as EAX reverb, dolby digital output on some devices, and better 3d sound virtualization using headphones.

Note that [FMOD\\_OPENRAW](#), [FMOD\\_OPENMEMORY](#), [FMOD\\_OPENMEMORY\\_POINT](#) and [FMOD\\_OPENUSER](#) will not work here without the exinfo structure present, as more information is needed.

Use [FMOD\\_NONBLOCKING](#) to have the sound open or load in the background. You can use [Sound::getOpenState](#) to determine if it has finished loading / opening or not. While it is loading (not ready), sound functions are not accessible for that sound.

To account for slow devices or computers that might cause buffer underrun (skipping/stuttering/repeating blocks of audio), use [System::setStreamBufferSize](#).

Note that [FMOD\\_CREATESAMPLE](#) will be ignored, overridden by this function because this is simply a wrapper to [System::createSound](#) that provides the [FMOD\\_CREATESTREAM](#) flag. The [FMOD\\_CREATESTREAM](#) flag overrides [FMOD\\_CREATESAMPLE](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MODE](#)
- [FMOD\\_CREATESOUNDEXINFO](#)
- [Sound::getOpenState](#)
- [System::setStreamBufferSize](#)
- [System::createSound](#)
- [Channel::setPan](#)

# System::get3DListenerAttributes

This retrieves the position, velocity and orientation of the specified 3D sound listener.?

## Syntax

```
FMOD_RESULT System::get3DListenerAttributes (
    int listener,
    FMOD_VECTOR * pos,
    FMOD_VECTOR * vel,
    FMOD_VECTOR * forward,
    FMOD_VECTOR * up
);
```

## Parameters

*listener*

Listener ID in a multi-listener environment. Specify 0 if there is only 1 listener.

*pos*

Address of a variable that receives the position of the listener in world space, measured in distance units. Optional. Specify 0 or NULL to ignore.

*vel*

Address of a variable that receives the velocity of the listener measured in distance units **per second**. Optional. Specify 0 or NULL to ignore.

*forward*

Address of a variable that receives the forwards orientation of the listener. Optional. Specify 0 or NULL to ignore.

*up*

Address of a variable that receives the upwards orientation of the listener. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

**See Also**

- [\\_System::set3DListenerAttributes](#)
- [\\_FMOD\\_VECTOR](#)

Version 4.12.03 Built on Feb 18, 2008

# System::get3DNumListeners

Retrieves the number of 3D listeners.?

## Syntax

```
FMOD_RESULT System::get3DNumListeners (
    int * numlisteners
);
```

## Parameters

*numlisteners*

Address of a variable that receives the current number of 3D listeners in the 3D scene.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::set3DNumListeners](#)

# System::get3DSettings

Retrieves the global doppler scale, distance factor and rolloff scale for all 3D sound in FMOD.?

## Syntax

```
FMOD_RESULT System::get3DSettings (
    float * dopplerscale,
    float * distancefactor,
    float * rolloffscale
);
```

## Parameters

*dopplerscale*

Address of a variable that receives the scaling factor for doppler shift. Optional. Specify 0 or NULL to ignore.

*distancefactor*

Address of a variable that receives the relative distance factor to FMOD's units. Optional. Specify 0 or NULL to ignore.

*rolloffscale*

Address of a variable that receives the scaling factor for 3D sound rolloff or attenuation. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::set3DSettings](#)

# System::get3DSpeakerPosition

Retrieves the current speaker position information for the selected speaker.?

## Syntax

```
FMOD_RESULT System::get3DSpeakerPosition(
    FMOD_SPEAKER speaker,
    float * x,
    float * y,
    bool * active
);
```

## Parameters

*speaker*

The selected speaker of interest to return the x and y position.

*x*

Address of a variable that receives the 2D X position relative to the listener. Optional. Specify 0 or NULL to ignore.

*y*

Address of a variable that receives the 2D Y position relative to the listener. Optional. Specify 0 or NULL to ignore.

*active*

Address of a variable that receives the active state of a speaker.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

See the [System::set3DSpeakerPosition](#) for more information on speaker positioning.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::set3DSpeakerPosition](#)
- [FMOD\\_SPEAKERMODE](#)
- [FMOD\\_SPEAKER](#)

# System::getAdvancedSettings

Retrieves the advanced settings value set for the system object.?

## Syntax

```
FMOD_RESULT System::getAdvancedSettings (
    FMOD_ADVANCEDSETTINGS * settings
);
```

## Parameters

*settings*

Address of a variable to receive the contents of the [FMOD\\_ADVANCEDSETTINGS](#) structure specified by the user.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_ADVANCEDSETTINGS](#)
- [System::setAdvancedSettings](#)



# System::getCDROMDriveName

Gets information on the selected cdrom drive.?

## Syntax

```
FMOD_RESULT System::getCDROMDriveName (
    int drive,
    char* driveName,
    int driveNameLen,
    char* scsiName,
    int scsiNameLen,
    char* devName,
    int devNameLen
);
```

## Parameters

*drive*

The enumerated number of the CDROM drive to query. 0 based.

*drivename*

Address of a variable that receives the name of the drive letter or name depending on the operating system.

*drivenamelen*

Length in bytes of the target buffer to receive the string.

*scsiname*

Address of a variable that receives the SCSI address of the drive. This could also be used to pass to [System::createSound](#), or just used for information purposes.

*scsinamelen*

Length in bytes of the target buffer to receive the string.

*devicename*

Address of a variable that receives the name of the physical device. This is usually a string defined by the manufacturer. It also contains the drive's vendor ID, product ID and version number.

*devicenamelen*

Length in bytes of the target buffer to receive the string.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Enumerate CDROM drives by finding out how many there are with [System::getNumCDROMDrives](#).

## Platforms Supported

Win32, Win64, PlayStation 3

## See Also

- [System::getNumCDROMDrives](#)
- [System::createSound](#)

# System::getCPUUsage

Retrieves in percent of CPU time - the amount of cpu usage that FMOD is taking for streaming/mixing and [System::update](#) combined.?

## Syntax

```
FMOD_RESULT System::getCPUUsage (
    float * dsp,
    float * stream,
    float * update,
    float * total
);
```

## Parameters

*dsp*

Address of a variable that receives the current dsp mixing engine cpu usage. Result will be from 0 to 100.0f. Optional. Specify 0 or NULL to ignore.

*stream*

Address of a variable that receives the current streaming engine cpu usage. Result will be from 0 to 100.0f. Optional. Specify 0 or NULL to ignore.

*update*

Address of a variable that receives the current [System::update](#) cpu usage. Result will be from 0 to 100.0f. Optional. Specify 0 or NULL to ignore.

*total*

Address of a variable that receives the current total cpu usage. Result will be from 0 to 100.0f. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This value is slightly smoothed to provide more stable readout (and to round off spikes that occur due to multitasking/operating system issues).

NOTE! On ps3 and xbox360, the dsp and stream figures are NOT main cpu/main thread usage. On PS3 this is the percentage of SPU being used. On Xbox 360 it is the percentage of a hardware thread being used which is on a totally different CPU than the main one.

Do not be alarmed if the usage for these platforms reaches over 50%, this is normal and should be ignored if you are playing a lot of compressed sounds and are using effects. The only value on the main cpu / main thread to take note of here that will impact your framerate is the update value, and this is typically very low (ie less than 1%).

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_System:update](#)

# System::getChannel

Retrieves a handle to a channel by ID.?

## Syntax

```
FMOD_RESULT System::getChannel(  
    int channelid,  
    FMOD_CHANNEL** channel  
);
```

## Parameters

*channelid*

Index in the FMOD channel pool. Specify a channel number from 0 to the 'maxchannels' value specified in [System::init](#) minus 1.

*channel*

Address of a variable that receives a pointer to the requested channel.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is mainly for getting handles to existing (playing) channels and setting their attributes.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::playSound](#)
- [System::init](#)

# System::getChannelsPlaying

Retrieves the number of currently playing channels.?

## Syntax

```
FMOD_RESULTSystem::getChannelsPlaying(  
    int * channels  
);
```

## Parameters

*channels*

Address of a variable that receives the number of currently playing channels.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

# System::getDSPBufferSize

Retrieves the buffer size settings for the FMOD software mixing engine.?

## Syntax

```
FMOD_RESULT System::getDSPBufferSize (
    unsigned int *    bufferlength,
    int *             numbuffers
);
```

## Parameters

*bufferlength*

Address of a variable that receives the mixer engine block size in samples. Default = 1024. (milliseconds = 1024 at 48khz = 1024 / 48000 \* 1000 = 10.66ms). This means the mixer updates every 21.3ms. Optional. Specify 0 or NULL to ignore.

*numbuffers*

Address of a variable that receives the mixer engine number of buffers used. Default = 4. To get the total buffersize multiply the bufferlength by the numbuffers value. By default this would be 4\*1024. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

See documentation on [System::setDSPBufferSize](#) for more information about these values.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setDSPBufferSize](#)





# System::getDSPHead

Returns a pointer to the head DSP unit of the DSP network. This unit is the closest unit to the soundcard and all sound data comes through this unit.

## Syntax

```
FMOD_RESULT System::getDSPHead(
    FMOD_DSP** dsp
);
```

## Parameters

*dsp*

Address of a variable that receives the pointer to the head DSP unit.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use this unit if you wish to connect custom DSP units to the output or filter the global mix by inserting filter units between this one and the incoming channel mixer unit.

Read the tutorial on DSP if you wish to know more about this. It is not recommended using this if you do not understand how the FMOD Ex DSP network is connected.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getDSPHead](#)
- [ChannelGroup::getDSPHead](#)

# System::getDriver

Returns the currently selected driver number. Drivers are enumerated when selecting a driver with [System::setDriver](#) or other driver related functions such as [System::getNumDrivers](#) or [System::getDriverInfo](#)?

## Syntax

```
FMOD_RESULT System::getDriver(  
    int * driver  
);
```

## Parameters

*driver*

Address of a variable that receives the currently selected driver ID. 0 = primary or main sound device as selected by the operating system settings.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setDriver](#)
- [System::getNumDrivers](#)
- [System::getDriverInfo](#)

# System::getDriverCaps

Returns information on capabilities of the current output mode for the selected sound device.?

## Syntax

```
FMOD_RESULT System::getDriverCaps (
    int id,
    FMOD_CAPS * caps,
    int * minfrequency,
    int * maxfrequency,
    FMOD_SPEAKERMODE * controlpanel_speaker_mode
);
```

## Parameters

*id*

Enumerated driver ID. This must be in a valid range delimited by [System::getNumDrivers](#).

*caps*

Address of a variable that receives the capabilities of the device. Optional. Specify 0 or NULL to ignore.

*minfrequency*

Address of a variable that receives the minimum frequency allowed with sounds created with [FMOD\\_HARDWARE](#). If [Channel::setFrequency](#) is used FMOD will clamp the frequency to this minimum. Optional. Specify 0 or NULL to ignore.

*maxfrequency*

Address of a variable that receives the maximum frequency allowed with sounds created with [FMOD\\_HARDWARE](#). If [Channel::setFrequency](#) is used FMOD will clamp the frequency to this maximum. Optional. Specify 0 or NULL to ignore.

*controlpanelspeakermode*

Address of a variable that receives the speaker mode set by the operating system control panel. Use this to pass to [System::setSpeakerMode](#) if you want to set up FMOD's software mixing engine to match. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function cannot be called after FMOD is already activated with [System::init](#).  
It must be called before [System::init](#), or after [System::close](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_CAPS](#)
- [System::init](#)
- [System::close](#)
- [System::getNumDrivers](#)
- [System::getHardwareChannels](#)
- [System::setSpeakerMode](#)
- [Channel::setFrequency](#)

# System::getDriverInfo

Retrieves identification information about a sound device specified by its index, and specific to the output mode set with [System::setOutput](#)?

## Syntax

```
FMOD_RESULT System::getDriverInfo (
    int id,
    char * name,
    int nameLen,
    FMOD_GUID * guid
);
```

## Parameters

*id*

Index of the sound driver device. The total number of devices can be found with [System::getNumDrivers](#).

*name*

Address of a variable that receives the name of the device. Optional. Specify 0 or NULL to ignore.

*namelen*

Length in bytes of the target buffer to receive the string. Required if name parameter is not NULL.

*guid*

Address of a variable that receives the GUID that uniquely identifies the device. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getNumDrivers](#)
- [System::setOutput](#)



# System::getGeometrySettings

Retrieves the maximum world size for the geometry engine.?

## Syntax

```
FMOD_RESULT System::getGeometrySettings (
    float * maxWorldSize
);
```

## Parameters

*maxWorldSize*

Pointer to a float to receive the maximum world size.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setGeometrySettings](#)

# System::getHardwareChannels

Returns the number of available hardware mixed 2d and 3d channels.?

## Syntax

```
FMOD_RESULT System::getHardwareChannels (
    int *    num2d,
    int *    num3d,
    int *    total
);
```

## Parameters

*num2d*

Address of a variable that receives the number of available hardware mixed 3d channels. Optional. Specify 0 or NULL to ignore.

*num3d*

Address of a variable that receives the number of available hardware mixed 2d channels. Optional. Specify 0 or NULL to ignore.

*total*

Address of a variable that receives the total number of available hardware mixed channels. Usually total = num3d + num2d, but on some platforms like PS2 and GameCube, 2D and 3D voices share the same channel pool so total, num2d and num3d will all be the same number. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If total doesn't equal num3d + num2d, it usually means the 2d and 3d hardware voices share the same actual hardware voice pool.

For example if it said 32 for each value, then if 30 3d voices were playing, then only 2 voices total would be available, for 2d or 3d playback. They are not separate.

NOTE: If this is called before [System::init](#), you must call [System::getDriverCaps](#) first.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable,



PlayStation 3, Wii, Solaris

#### **See Also**

- [\\_System:init](#)
- [\\_System:getChannelsPlaying](#)
- [\\_System:setHardwareChannels](#)
- [\\_System:getDriverCaps](#)

Version 4.12.03 Built on Feb 18, 2008

# System::getMasterChannelGroup

Retrieves a handle to the internal master channel group. This is the default channel group that all channels play on. This channel group can be used to do things like set the master volume for all playing sounds. See the ChannelGroup API for more functionality.

## Syntax

```
FMOD_RESULT System::getMasterChannelGroup(
    FMOD_CHANNELGROUP **channelgroup
);
```

## Parameters

*channelgroup*

Address of a variable that receives a pointer to the master System object channel group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createChannelGroup](#)
- [ChannelGroup::setVolume](#)
- [ChannelGroup::getVolume](#)

# System::getMasterSoundGroup

Retrieves the default sound group, where all sounds are placed when they are created.?

## Syntax

```
FMOD_RESULT System::getMasterSoundGroup(
    FMOD_SOUNDGROUP ** soundgroup
);
```

## Parameters

*soundgroup*

Address of a pointer to a SoundGroup object to receive the master sound group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a user based soundgroup is deleted/released, the sounds will be put back into this sound group.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::release](#)
- [SoundGroup::getSystemObject](#)
- [SoundGroup::setMaxAudible](#)
- [SoundGroup::getMaxAudible](#)
- [SoundGroup::getName](#)
- [SoundGroup::getNumSounds](#)
- [SoundGroup::getSound](#)
- [SoundGroup::getNumPlaying](#)
- [SoundGroup::setUserData](#)
- [SoundGroup::getUserData](#)



# System::getNetworkProxy

Retrieves the URL of the proxy server used in internet streaming.?

## Syntax

```
FMOD_RESULT System::getNetworkProxy (
    char * proxy,
    int proxylen
);
```

## Parameters

*proxy*

Address of a variable that receives the proxy server URL.

*proxylen*

Size of the buffer in bytes to receive the string.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, PlayStation 3

## See Also

- [System::setNetworkProxy](#)

# System::getNetworkTimeout

Retrieve the timeout value for network streams?

## Syntax

```
FMOD_RESULTSystem::getNetworkTimeout(  
    int * timeout  
);
```

## Parameters

*timeout*

The timeout value in ms.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, PlayStation 3

# System::getNumCDROMDrives

Retrieves the number of available CDROM drives on the user's machine.?

## Syntax

```
FMOD_RESULT System::getNumCDROMDrives (
    int * numdrives
);
```

## Parameters

*numdrives*

Address of a variable that receives the number of CDROM drives.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getCDROMDriveName](#)

# System::getNumDrivers

Retrieves the number of soundcard devices on the machine, specific to the output mode set with [System::setOutput](#)?

## Syntax

```
FMOD_RESULT System::getNumDrivers (
    int * numdrivers
);
```

## Parameters

*numdrivers*

Address of a variable that receives the number of output drivers.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If [System::setOutput](#) is not called it will return the number of drivers available for the default output type. Use this for enumerating sound devices. Use [System::getDriverInfo](#) to get the device's name.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getDriver](#)
- [System::getDriverInfo](#)
- [System::setOutput](#)
- [System::getOutput](#)



# System::getNumPlugins

Retrieves the number of available plugins loaded into FMOD at the current time.?

## Syntax

```
FMOD_RESULT System::getNumPlugins (
    FMOD_PLUGINTYPE pluginType,
    int * numPlugins
);
```

## Parameters

*pluginType*

Specify the type of plugin type such as [FMOD\\_PLUGINTYPE\\_OUTPUT](#), [FMOD\\_PLUGINTYPE\\_CODEC](#) or [FMOD\\_PLUGINTYPE\\_DSP](#).

*numPlugins*

Address of a variable that receives the number of available plugins for the selected type.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_PLUGINTYPE](#)

# System::getOutput

Retrieves the current output system FMOD is using to address the hardware.?

## Syntax

```
FMOD_RESULT System::getOutput(
    FMOD_OUTPUTTYPE *output
);
```

## Parameters

*output*

Address of a variable that receives the current output type.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_OUTPUTTYPE](#)

# System::getOutputByPlugin

Returns the currently selected output as an id in the list of output plugins.?

## Syntax

```
FMOD_RESULT System::getOutputByPlugin(  
    int * index  
);
```

## Parameters

*index*

Address of a variable that receives the currently selected output as an index in a plugin list.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::getNumPlugins](#)
- [\\_System::setOutputByPlugin](#)
- [\\_System::setOutput](#)

# System::getOutputHandle

Retrieves a pointer to the system level output device module. This means a pointer to a DirectX "LPGDIRECTIONSOUND", or a WINMM handle, or with something like with [FMOD\\_OUTPUTTYPE\\_NOSOUND](#) output, the handle will be null or 0.?

## Syntax

```
FMOD_RESULT System::getOutputHandle (
    void ** handle
);
```

## Parameters

*handle*

Address of a variable that receives the handle to the output mode's native hardware API object (see remarks).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Must be called after [System::init](#).

Cast the resulting pointer depending on what output system pointer you are after.

**FMOD\_OUTPUTTYPE\_WINMM** Pointer to type HWAVEOUT is returned.

**FMOD\_OUTPUTTYPE\_DSOUND** Pointer to type DIRECTSOUND is returned.

**FMOD\_OUTPUTTYPE\_ASIO** Pointer to type AsioDrivers is returned.

**FMOD\_OUTPUTTYPE\_OSS** File handle is returned, (cast to int).

**FMOD\_OUTPUTTYPE\_ESD** Handle of type int is returned, as returned by so\_esd\_open\_sound (cast to int).

**FMOD\_OUTPUTTYPE\_ALSA** Pointer to type snd\_pcm\_t is returned.

**FMOD\_OUTPUTTYPE\_MAC** Handle of type SndChannelPtr is returned.

**FMOD\_OUTPUTTYPE\_Xbox** Pointer to type DIRECTSOUND is returned.

**FMOD\_OUTPUTTYPE\_PS2** NULL / 0 is returned.

**FMOD\_OUTPUTTYPE\_GC** NULL / 0 is returned.

**FMOD\_OUTPUTTYPE\_NOSOUND** NULL / 0 is returned.

**FMOD\_OUTPUTTYPE\_WAVWRITER** NULL / 0 is returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_OUTPUTTYPE](#)
- [System:setOutput](#)
- [System:init](#)

Version 4.12.03 Built on Feb 18, 2008

# System::getPluginInfo

Retrieves information to display for the selected plugin.?

## Syntax

```
FMOD_RESULT System::getPluginInfo (
    FMOD_PLUGINTYPE pluginType,
    int index,
    char * name,
    int nameLen,
    unsigned int * version
);
```

## Parameters

*pluginType*

Specify the type of plugin type such as [FMOD\\_PLUGINTYPE\\_OUTPUT](#), [FMOD\\_PLUGINTYPE\\_CODEC](#) or [FMOD\\_PLUGINTYPE\\_DSP](#).

*index*

Index into the enumerated list of output plugins.

*name*

Address of a variable that receives the name of the plugin.

*namelen*

Length in bytes of the target buffer to receive the string.

*version*

Version number set by the plugin.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::getNumPlugins](#)

Version 4.12.03 Built on Feb 18, 2008

# System::getRecordDriver

Retrieves the currently selected recording driver, usually set with [System::setRecordDriver](#).

## Syntax

```
FMOD_RESULT System::getRecordDriver(  
    int * driver  
);
```

## Parameters

*driver*

Address of a variable to receive the currently selected recording driver.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The number of drivers available can be retrieved with [System::getNumDrivers](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [System::setRecordDriver](#)
- [System::getNumDrivers](#)



# System::getRecordDriverCaps

Returns information on capabilities of the current output mode for the selected recording sound device.?

## Syntax

```
FMOD_RESULT System::getRecordDriverCaps (
    int id,
    FMOD_CAPS * caps,
    int * minfrequency,
    int * maxfrequency
);
```

## Parameters

*id*

Enumerated driver ID. This must be in a valid range delimited by [System::getRecordNumDrivers](#).

*caps*

Address of a variable that receives the capabilities of the device. Optional. Specify 0 or NULL to ignore.

*minfrequency*

Address of a variable that receives the minimum frequency allowed for sounds used with recording. Optional. Specify 0 or NULL to ignore.

*maxfrequency*

Address of a variable that receives the maximum frequency allowed for sounds used with recording. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [FMOD\\_CAPS](#)
- [System::getRecordNumDrivers](#)



# System::getRecordDriverInfo

Retrieves identification information about a sound device specified by its index, and specific to the output mode set with [System::setOutput](#)?

## Syntax

```
FMOD_RESULT System::getRecordDriverInfo (
    int id,
    char * name,
    int nameLen,
    FMOD_GUID * guid
);
```

## Parameters

*id*

Index into the enumerated list of record devices up to the value returned by [System::getRecordNumDrivers](#).

*name*

Address of a variable that receives the name of the recording device. Optional. Specify 0 or NULL to ignore.

*namelen*

Length in bytes of the target buffer to receive the string. Required if name parameter is not NULL.

*guid*

Address of a variable that receives the GUID that uniquely identifies the device. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [System::setOutput](#)
- [System::getRecordNumDrivers](#)



# System::getRecordNumDrivers

Retrieves the number of recording devices available for this output mode. Use this to enumerate all recording devices possible so that the user can select one.?

## Syntax

```
FMOD_RESULT System::getRecordNumDrivers (
    int * numDrivers
);
```

## Parameters

*numDrivers*

Address of a variable that receives the number of recording drivers available for this output mode.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [System::GetRecordDriverInfo](#)

# System::getRecordPosition

Retrieves the current recording position of the record buffer in PCM samples.?

## Syntax

```
FMOD_RESULT System::getRecordPosition(  
    unsigned int * position  
);
```

## Parameters

*position*

Address of a variable to receive the current recording position in PCM samples.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

# System::getReverbAmbientProperties

Retrieves the default reverb environment for the virtual reverb system.

## Syntax

```
FMOD_RESULT System::getReverbAmbientProperties (
    FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of a pointer to a [FMOD\\_REVERB\\_PROPERTIES](#) to receive the settings for the current ambient reverb setting.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

By default the ambient reverb is set to 'off'. This is the same as FMOD\_REVERB\_PRESET\_OFF.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [System::setReverbAmbientProperties](#)
- [System::createReverb](#)

# System::getReverbProperties

Retrieves the current reverb environment.?

## Syntax

```
FMOD_RESULT System::getReverbProperties (
    FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of a variable that receives the current reverb environment description.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setReverbProperties](#)
- [Channel::setReverbProperties](#)
- [Channel::getReverbProperties](#)



# System::getSoftwareChannels

Retrieves the maximum number of software mixed channels possible. Software mixed voices are used by sounds loaded with [FMOD\\_SOFTWARE](#).

## Syntax

```
FMOD_RESULT System::getSoftwareChannels (
    int * numsoftwarechannels
);
```

## Parameters

*numsoftwarechannels*

Address of a variable that receives the current maximum number of software voices available.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setSoftwareChannels](#)

# System::getSoftwareFormat

Retrieves the output format for the software mixer.?

## Syntax

```
FMOD_RESULT System::getSoftwareFormat(  
    int * samplerate,  
    FMOD_SOUND_FORMAT * format,  
    int * numoutputchannels,  
    int * maxinputchannels,  
    FMOD_RESAMPLE_METHOD * resamplemethod,  
    int * bits  
);
```

## Parameters

*samplerate*

Address of a variable that receives the mixer's output rate. Optional. Specify 0 or NULL to ignore.

*format*

Address of a variable that receives the mixer's output format. Optional. Specify 0 or NULL to ignore.

*numoutputchannels*

Address of a variable that receives the number of output channels to initialize the mixer to, for example 1 = mono, 2 = stereo. 8 is the maximum for soundcards that can handle it. Optional. Specify 0 or NULL to ignore.

*maxinputchannels*

Address of a variable that receives the maximum channel depth on sounds that are loadable or creatable. Specify 0 or NULL to ignore.

*resamplemethod*

Address of a variable that receives the current resampling (frequency conversion) method for software mixed sounds. Specify 0 or NULL to ignore.

*bits*

Address of a variable that receives the number of bits per sample. Useful for byte->sample conversions. for example [FMOD\\_SOUND\\_FORMAT\\_PCM16](#) is 16. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note that the settings returned here may differ from the settings provided by the user with [System::setSoftwareFormat](#). This is because the driver may have changed it because it will not initialize to anything else.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::setSoftwareFormat](#)
- [\\_FMOD\\_SOUND\\_FORMAT](#)
- [\\_FMOD\\_DSP\\_RESAMPLER](#)

# System::getSoundRAM

Retrieves the amount of dedicated sound ram available if the platform supports it. Currently only support on GameCube.

?Most platforms use main ram to store audio data, so this function usually isn't necessary.?

## Syntax

```
FMOD_RESULT System::getSoundRAM (
    int * currentAllocated,
    int * maxAllocated,
    int * total
);
```

## Parameters

*currentAllocated*

Address of a variable that receives the currently allocated sound ram memory at time of call. Optional. Specify 0 or NULL to ignore.

*maxAllocated*

Address of a variable that receives the maximum allocated sound ram memory since System::init. Optional. Specify 0 or NULL to ignore.

*total*

Address of a variable that receives the total amount of sound ram available on this device.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

In the future Will support PlayStation 2 (SPU2 RAM), and Creative X-Fi sound card when it comes out as it has dedicated sound ram.

## Platforms Supported

GameCube, PlayStation 3

## See Also

- [\\_Memory\\_GetStats](#)

Version 4.12.03 Built on Feb 18, 2008

# System::getSpeakerMode

Retrieves the current speaker mode.?

## Syntax

```
FMOD_RESULT System::getSpeakerMode (
    FMOD_SPEAKERMODE * speakerMode
);
```

## Parameters

*speakerMode*

Address of a variable that receives the current speaker mode.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setSpeakerMode](#)
- [FMOD\\_SPEAKERMODE](#)

# System::getSpectrum

Retrieves the spectrum from the currently playing output signal.?

## Syntax

```
FMOD_RESULT System::getSpectrum(  
    float * spectrumarray,  
    int numvalues,  
    int channeloffset,  
    FMOD_DSP_FFT_WINDOW windowtype  
);
```

## Parameters

*spectrumarray*

Address of a variable that receives the spectrum data. This is an array of floating point values. Data will range is 0.0 to 1.0. Decibels =  $10.0f * (\text{float})\log_{10}(\text{val}) * 2.0f$ ; See remarks for what the data represents.

*numvalues*

Size of array in floating point values being passed to the function. Must be a power of 2. (ie 128/256/512 etc). Min = 64. Max = 8192.

*channeloffset*

Channel of the signal to analyze. If the signal is multichannel (such as a stereo output), then this value represents which channel to analyze. On a stereo signal 0 = left, 1 = right.

*windowtype*

"Pre-FFT" window method. This filters the PCM data before entering the spectrum analyzer to reduce transient frequency error for more accurate results. See [FMOD\\_DSP\\_FFT\\_WINDOW](#) for different types of fft window techniques possible and for a more detailed explanation.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The larger the numvalues, the more CPU the FFT will take. Choose the right value to trade off between accuracy / speed.

The larger the numvalues, the more 'lag' the spectrum will seem to inherit. This is because the FFT window size stretches the analysis back in time to what was already played. For example if the window size happened to be 44100 and the output rate was 44100 it would be analyzing the past second of data, and giving you the average spectrum over that time period.

If you are not displaying the result in dB, then the data may seem smaller than it should be. To display it you may want to normalize the data - that is, find the maximum value in the resulting spectrum, and scale all values in the array by  $1 / \text{max}$ . (ie if the max was 0.5f, then it would become 1).

To get the spectrum for both channels of a stereo signal, call this function twice, once with `channeloffset = 0`, and again with `channeloffset = 1`. Then add the spectrums together and divide by 2 to get the average spectrum for both channels.

#### What the data represents.

To work out what each entry in the array represents, use this formula

$$\text{entry\_hz} = \text{output\_rate} / 2 / \text{numvalues}$$

The array represents amplitudes of each frequency band from 0hz to the nyquist rate. The nyquist rate is equal to the output rate divided by 2.

For example when FMOD is set to 44100hz output, the range of represented frequencies will be 0hz to 22049hz, a total of 22050hz represented.

If in the same example, 1024 was passed to this function as the numvalues, each entry's contribution would be as follows.

$$\text{entry\_hz} = (44100 / 2 / 1024)$$
$$\text{entry\_hz} = 21.53 \text{ hz}$$

**Note:** This function only displays data for sounds playing that were created with [FMOD\\_SOFTWARE](#). [FMOD\\_HARDWARE](#) based sounds are played using the sound card driver and are not accessible.

### Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

### See Also

- [FMOD\\_DSP\\_FFT\\_WINDOW](#)
- [Channel::getSpectrum](#)
- [ChannelGroup::getSpectrum](#)
- [System::getWaveData](#)



# System::getStreamBufferSize

Returns the current internal buffersize settings for streamable sounds.?

## Syntax

```
FMOD_RESULT System::getStreamBufferSize (
    unsigned int * filebuffersize,
    FMOD_TIMEUNIT * filebuffersizetype
);
```

## Parameters

*filebuffersize*

Address of a variable that receives the current stream file buffer size setting. Default is 16384 ([FMOD\\_TIMEUNIT\\_RAWBYTES](#)). Optional. Specify 0 or NULL to ignore.

*filebuffersizetype*

Address of a variable that receives the type of unit for the current stream file buffer size setting. Can be [FMOD\\_TIMEUNIT\\_MS](#), [FMOD\\_TIMEUNIT\\_PCM](#), [FMOD\\_TIMEUNIT\\_PCMBYTES](#) or [FMOD\\_TIMEUNIT\\_RAWBYTES](#). Default is [FMOD\\_TIMEUNIT\\_RAWBYTES](#). Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_TIMEUNIT](#)
- [System::setStreamBufferSize](#)

# System::getUserData

Retrieves the user value that that was set by calling the [System::setUserData](#) function.?

## Syntax

```
FMOD_RESULT System::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [System::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setUserData](#)

# System::getVersion

Returns the current version of FMOD Ex being used.?

## Syntax

```
FMOD_RESULT System::getVersion(
    unsigned int * version
);
```

## Parameters

*version*

Address of a variable that receives the current FMOD Ex version.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The version is a 32bit hexadecimal value formatted as 16:8:8, with the upper 16bits being the major version, the middle 8bits being the minor version and the bottom 8bits being the development version. For example a value of 00040106h is equal to 4.01.06.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::init](#)

# System::getWaveData

Retrieves a pointer to a block of PCM data that represents the currently playing audio mix.

?This function is useful for a very easy way to plot an oscilloscope.

?

## Syntax

```
FMOD_RESULT System::getWaveData (
    float * wavearray,
    int numvalues,
    int channeloffset
);
```

## Parameters

*wavearray*

Address of a variable that receives the currently playing waveform data. This is an array of floating point values.

*numvalues*

Number of floats to write to the array. Maximum value = 16384.

*channeloffset*

Offset into multichannel data. For mono output use 0. Stereo output will use 0 = left, 1 = right. More than stereo output - use the appropriate index.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This is the actual resampled, filtered and volume scaled data of the final output, at the time this function is called.

Do not use this function to try and display the whole waveform of the sound, as this is more of a 'snapshot' of the current waveform at the time it is called, and could return the same data if it is called very quickly in succession. See the DSP API to capture a continual stream of wave data as it plays, or see [Sound::lock](#) / [Sound::unlock](#) if you want to simply display the waveform of a sound.

This function allows retrieval of left and right data for a stereo sound individually. To combine them into one signal, simply add the entries of each separate buffer together and then divide them by 2.

**Note:** This function only displays data for sounds playing that were created with [FMOD\\_SOFTWARE](#). [FMOD\\_HARDWARE](#) based sounds are played using the sound card driver and are not accessible.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::getSpectrum](#)
- [\\_Channel::getWaveData](#)
- [\\_ChannelGroup::getWaveData](#)
- [\\_Sound::lock](#)
- [\\_Sound::unlock](#)

# System::init

Initializes the system object, and the sound device. This has to be called at the start of the user's program.

**You must create a system object with FMOD::System\_create.**

## Syntax

```
FMOD_RESULT System::init(  
    int maxchannels,  
    FMOD_INIT_FLAGS flags,  
    void* extradriverdata  
);
```

## Parameters

*maxchannels*

The maximum number of channels to be used in FMOD. They are also called 'virtual channels' as you can play as many of these as you want, even if you only have a small number of hardware or software voices. See remarks for more.

*flags*

See [FMOD\\_INITFLAGS](#). This can be a selection of flags bitwise OR'ed together to change the behaviour of FMOD at initialization time.

*extradriverdata*

Driver specific data that can be passed to the output plugin. For example the filename for the wav writer plugin. See [FMOD\\_OUTPUTTYPE](#) for what each output mode might take here. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### Virtual channels.

These types of voices are the ones you work with using the FMOD::Channel API.

The advantage of virtual channels are, unlike older versions of FMOD, you can now play as many sounds as you like without fear of ever running out of voices, or playsound failing.

You can also avoid 'channel stealing' if you specify enough virtual voices.

As an example, you can play 1000 sounds at once, even on a 32 channel soundcard.

FMOD will only play the most important/closest/loudest (determined by volume/distance/geometry and priority settings) voices, and the other 968 voices will be virtualized without expense to the CPU. The voice's cursor positions are updated.

When the priority of sounds change or emulated sounds get louder than audible ones, they will swap the actual voice

resource over (ie hardware or software buffer) and play the voice from its correct position in time as it should be heard.

What this means is you can play all 1000 sounds, if they are scattered around the game world, and as you move around the world you will hear the closest or most important 32, and they will automatically swap in and out as you move.

Currently the maximum channel limit is 4093.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [FMOD\\_INITELAGS](#)
- [System::close](#)
- [System\\_Create](#)
- [FMOD\\_OUTPUTTYPE](#)

# System::isRecording

Retrieves the state of the FMOD recording API, ie if it is currently recording or not.?

## Syntax

```
FMOD_RESULT System::isRecording (
    bool * recording
);
```

## Parameters

*recording*

Address of a variable to receive the current recording state. True or non zero if the FMOD recording api is currently in the middle of recording, false or zero if the recording api is stopped / not recording.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Recording can be started with [System::recordStart](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [System::recordStart](#)
- [System::recordStop](#)



# System::loadGeometry

Creates a geometry object from a block of memory which contains pre-saved geometry data, saved by [Geometry::save](#)?

## Syntax

```
FMOD_RESULT System::loadGeometry (
    const void * data,
    int datasize,
    FMOD::Geometry ** geometry
);
```

## Parameters

*data*

Address of data containing pre-saved geometry data.

*datasize*

Size of geometry data block in bytes.

*geometry*

Address of a variable to receive a newly created FMOD::Geometry object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::save](#)
- [System::createGeometry](#)

# System::loadPlugin

Loads an FMOD plugin. This could be a DSP, file format or output plugin.?

## Syntax

```
FMOD_RESULT System::loadPlugin(  
    const char * filename ,  
    FMOD_PLUGIN_TYPE * pluginType ,  
    int * index  
);
```

## Parameters

*filename*

Filename of the plugin to be loaded.

*pluginType*

Address of a variable that receives the type of plugin that has been loaded (if successful).

*index*

Index into the plugin list for that plugin type. Use this for DSP plugins created with [System::createDSPByIndex](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Once the plugin is loaded, it can be enumerated and used.

For file format plugins, FMOD will automatically try to use them when [System::createSound](#) is used.

For DSP plugins, you can enumerate them with [System::getNumPlugins](#) and [System::getPluginInfo](#).

Plugins can be created for FMOD by the user. See the relevant section in the documentation on creating plugins.

The format of the plugin is dependant on the operating system.

On Win32 and Win64 the .dll format is used

On Linux, the .so format is used.

On Macintosh, the .shlib format is used?

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh.

## See Also

- [\\_System::setPluginPath](#)
- [\\_System::createSound](#)
- [\\_System::getNumPlugins](#)
- [\\_System::getPluginInfo](#)
- [\\_System::createDSPByIndex](#)

Version 4.12.03 Built on Feb 18, 2008

# System::lockDSP

Mutual exclusion function to lock the FMOD DSP engine (which runs asynchronously in another thread), so that it will not execute. If the FMOD DSP engine is already executing, this function will block until it has completed.

?The function may be used to synchronize DSP network operations carried out by the user.

?An example of using this function may be for when the user wants to construct a DSP sub-network, without the DSP engine executing in the background while the sub-network is still under construction.

?

## Syntax

```
FMOD_RESULT System::lockDSP();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Once the user no longer needs the DSP engine locked, it must be unlocked with [System::unlockDSP](#).

Note that the DSP engine should not be locked for a significant amount of time, otherwise inconsistency in the audio output may result. (audio skipping/stuttering).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::unlockDSP](#)

# System::playDSP

Plays a DSP unit object and its input network on a particular channel.

## Syntax

```
FMOD_RESULT System::playDSP(  
    FMOD_CHANNELINDEX channelid,  
    FMOD_DSP * dsp,  
    bool paused,  
    FMOD_CHANNEL ** channel  
);
```

## Parameters

*channelid*

Use the value [FMOD\\_CHANNEL\\_FREE](#) to get FMOD to pick a free channel. Otherwise specify a channel number from 0 to the 'maxchannels' value specified in [System::init](#) minus 1.

*dsp*

Pointer to the dsp unit to play. This is opened with [System::createDSP](#), [System::createDSPByType](#), [System::createDSPByIndex](#).

*paused*

True or false flag to specify whether to start the channel paused or not. Starting a channel paused allows the user to alter its attributes without it being audible, and unpausing with [Channel::setPaused](#) actually starts the dsp running.

*channel*

Address of a channel handle pointer that receives the newly playing channel. If [FMOD\\_CHANNEL\\_REUSE](#) is used, this can contain a previously used channel handle and FMOD will re-use it to play a dsp on.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When a dsp is played, it will use the dsp's default frequency, volume, pan, levels and priority.

A dsp defined as [FMOD\\_3D](#) will by default play at the position of the listener.

To change channel attributes before the dsp is audible, start the channel paused by setting the paused flag to true, and calling the relevant channel based functions. Following that, unpause the channel with [Channel::setPaused](#).

If [FMOD\\_CHANNEL\\_FREE](#) is used as the channel index, it will pick an arbitrary free channel and use channel management. (As described below).

If [FMOD\\_CHANNEL\\_REUSE](#) is used as the channel index, FMOD Ex will re-use the channel handle that is passed in as the 'channel' parameter. If NULL or 0 is passed in as the channel handle it will use the same logic as [FMOD\\_CHANNEL\\_FREE](#) and pick an arbitrary channel.

Channels are reference counted. If a channel is stolen by the FMOD priority system, then the handle to the stolen voice becomes invalid, and Channel based commands will not affect the new channel playing in its place.

If all channels are currently full playing a dsp or sound, FMOD will steal a channel with the lowest priority dsp or sound.

If more channels are playing than are currently available on the soundcard/sound device or software mixer, then FMOD will 'virtualize' the channel. This type of channel is not heard, but it is updated as if it was playing. When its priority becomes high enough or another sound stops that was using a real hardware/software channel, it will start playing from where it should be. This technique saves CPU time (thousands of sounds can be played at once without actually being mixed or taking up resources), and also removes the need for the user to manage voices themselves. An example of virtual channel usage is a dungeon with 100 torches burning, all with a looping crackling sound, but with a soundcard that only supports 32 hardware voices. If the 3D positions and priorities for each torch are set correctly, FMOD will play all 100 sounds without any 'out of channels' errors, and swap the real voices in and out according to which torches are closest in 3D space.

Priority for virtual channels can be changed in the sound's defaults, or at runtime with [Channel::setPriority](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_CHANNELINDEX](#)
- [System::createDSP](#)
- [System::createDSPByType](#)
- [System::createDSPByIndex](#)
- [Channel::setPaused](#)
- [Channel::setPriority](#)
- [DSP::setDefaults](#)
- [System::init](#)

# System::playSound

Plays a sound object on a particular channel.

## Syntax

```
FMOD_RESULT System::playSound(  
    FMOD_CHANNELINDEX channelid,  
    FMOD_SOUND * sound,  
    bool paused,  
    FMOD_CHANNEL ** channel  
);
```

## Parameters

*channelid*

Use the value [FMOD\\_CHANNEL\\_FREE](#) to get FMOD to pick a free channel. Otherwise specify a channel number from 0 to the 'maxchannels' value specified in [System::init](#) minus 1.

*sound*

Pointer to the sound to play. This is opened with [System::createSound](#).

*paused*

True or false flag to specify whether to start the channel paused or not. Starting a channel paused allows the user to alter its attributes without it being audible, and unpausing with [Channel::setPaused](#) actually starts the sound.

*channel*

Address of a channel handle pointer that receives the newly playing channel. If [FMOD\\_CHANNEL\\_REUSE](#) is used, this can contain a previously used channel handle and FMOD will re-use it to play a sound on.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When a sound is played, it will use the sound's default frequency, volume, pan, levels and priority.

A sound defined as [FMOD\\_3D](#) will by default play at the position of the listener.

To change channel attributes before the sound is audible, start the channel paused by setting the paused flag to true, and calling the relevant channel based functions. Following that, unpause the channel with [Channel::setPaused](#).

If [FMOD\\_CHANNEL\\_FREE](#) is used as the channel index, it will pick an arbitrary free channel and use channel

management. (As described below).

If [FMOD\\_CHANNEL\\_REUSE](#) is used as the channel index, FMOD Ex will re-use the channel handle that is passed in as the 'channel' parameter. If NULL or 0 is passed in as the channel handle it will use the same logic as [FMOD\\_CHANNEL\\_FREE](#) and pick an arbitrary channel.

Channels are reference counted. If a channel is stolen by the FMOD priority system, then the handle to the stolen voice becomes invalid, and Channel based commands will not affect the new sound playing in its place.

If all channels are currently full playing a sound, FMOD will steal a channel with the lowest priority sound.

If more channels are playing than are currently available on the soundcard/sound device or software mixer, then FMOD will 'virtualize' the channel. This type of channel is not heard, but it is updated as if it was playing. When its priority becomes high enough or another sound stops that was using a real hardware/software channel, it will start playing from where it should be. This technique saves CPU time (thousands of sounds can be played at once without actually being mixed or taking up resources), and also removes the need for the user to manage voices themselves. An example of virtual channel usage is a dungeon with 100 torches burning, all with a looping crackling sound, but with a soundcard that only supports 32 hardware voices. If the 3D positions and priorities for each torch are set correctly, FMOD will play all 100 sounds without any 'out of channels' errors, and swap the real voices in and out according to which torches are closest in 3D space.

Priority for virtual channels can be changed in the sound's defaults, or at runtime with [Channel::setPriority](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_CHANNELINDEX](#)
- [System::createSound](#)
- [Channel::setPaused](#)
- [Channel::setPriority](#)
- [Sound::setDefaults](#)
- [Sound::setVariations](#)
- [System::init](#)



# System::recordStart

Starts the recording engine recording to the specified recording sound.?

## Syntax

```
FMOD_RESULT Sys::recordStart(  
    FMOD::Sound * sound,  
    bool loop  
);
```

## Parameters

*sound*

User created sound for the user to record to.

*loop*

Boolean flag to tell the recording engine whether to continue recording to the provided sound from the start again, after it has reached the end. If this is set to true the data will be continually be overwritten once every loop. See remarks.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [\\_System::recordStop](#)

# System::recordStop

Stops the recording engine from recording to the specified recording sound.?

## Syntax

```
FMOD_RESULT Sys::recordStop;
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [\\_System::recordStart](#)

# System::release

Closes and frees a system object and its resources.?

## Syntax

```
FMOD_RESULT System::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function also calls [System::close](#), so calling close before this function is not necessary.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System\\_Create](#)
- [\\_System::init](#)
- [\\_System::close](#)

# System::set3DListenerAttributes

This updates the position, velocity and orientation of the specified 3D sound listener.?

## Syntax

```
FMOD_RESULT System::set3DListenerAttributes (
    int listener,
    const FMOD_VECTOR * pos,
    const FMOD_VECTOR * vel,
    const FMOD_VECTOR * forward,
    const FMOD_VECTOR * up
);
```

## Parameters

*listener*

Listener ID in a multi-listener environment. Specify 0 if there is only 1 listener.

*pos*

The position of the listener in world space, measured in distance units. You can specify 0 or NULL to not update the position.

*vel*

The velocity of the listener measured in distance units **per second**. You can specify 0 or NULL to not update the velocity of the listener.

*forward*

The forwards orientation of the listener. This vector must be of unit length and perpendicular to the up vector. You can specify 0 or NULL to not update the forwards orientation of the listener.

*up*

The upwards orientation of the listener. This vector must be of unit length and perpendicular to the forwards vector. You can specify 0 or NULL to not update the upwards orientation of the listener.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

By default, FMOD uses a left-handed co-ordinate system. This means +X is right, +Y is up, and +Z is forwards. To change this to a right-handed coordinate system, use [FMOD\\_INIT\\_3D\\_RIGHTHANDED](#). This means +X is

left, +Y is up, and +Z is forwards.

To map to another coordinate system, flip/negate and exchange these values.

Orientation vectors are expected to be of UNIT length. This means the magnitude of the vector should be 1.0.

A 'distance unit' is specified by [System::set3DSettings](#). By default this is set to meters which is a distance scale of 1.0.

Always remember to use **units per second**, *not* units per frame as this is a common mistake and will make the doppler effect sound wrong.

For example, Do not just use (pos - lastpos) from the last frame's data for velocity, as this is not correct. You need to time compensate it so it is given in units per **second**.

You could alter your pos - lastpos calculation to something like this.

```
vel = (pos - lastpos) / time_taken_since_last_frame_in_seconds.
```

I.e. at 60fps the formula would look like this vel = (pos-lastpos) / 0.0166667.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::get3DListenerAttributes](#)
- [FMOD\\_INITFLAGS](#)
- [System::set3DSettings](#)
- [System::get3DSettings](#)
- [FMOD\\_VECTOR](#)

# System::set3DNumListeners

Sets the number of 3D 'listeners' in the 3D sound scene. This function is useful mainly for split-screen game purposes.?

## Syntax

```
FMOD_RESULT System::set3DNumListeners (
    int numListeners
);
```

## Parameters

*numlisteners*

Number of listeners in the scene. Valid values are from 1-4 inclusive. Default = 1.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If the number of listeners is set to more than 1, then panning and doppler are turned off. All sound effects will be mono.

FMOD uses a 'closest sound to the listener' method to determine what should be heard in this case.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::get3DNumListeners](#)
- [System::set3DListenerAttributes](#)

# System::set3DRolloffCallback

When FMOD wants to calculate 3d volume for a channel, this callback can be used to override the internal volume calculation based on distance.?

## Syntax

```
FMOD_RESULT System::set3DRolloffCallback (
    FMOD_3D_CALLBACK callback
);
```

## Parameters

*callback*

Pointer to a C function of type [FMOD\\_3D\\_ROLLOFFCALLBACK](#), that is used to override the FMOD volume calculation. Default is 0 or NULL. Setting the callback to null will return 3d calculation back to FMOD.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function overrides [FMOD\\_3D\\_LOGROLLOFF](#), [FMOD\\_3D\\_LINEARROLLOFF](#), [FMOD\\_3D\\_CUSTOMROLLOFF](#). To allow FMOD to calculate the 3d volume again, use 0 or NULL as the callback.

NOTE: When using the event system, call [Channel::getUserData](#) from your [FMOD\\_3D\\_ROLLOFFCALLBACK](#) to get the event instance handle of the event that spawned the channel in question.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_3D\\_ROLLOFFCALLBACK](#)
- [System::set3DListenerAttributes](#)
- [System::get3DListenerAttributes](#)
- [Channel::getUserData](#)





# System::set3DSettings

Sets the global doppler scale, distance factor and log rolloff scale for all 3D sound in FMOD.?

## Syntax

```
FMOD_RESULT System::set3DSettings (
    float dopplerscale,
    float distancefactor,
    float rolloffscale
);
```

## Parameters

*dopplerscale*

Scaling factor for doppler shift. Default = 1.0.

*distancefactor*

Relative distance factor to FMOD's units. Default = 1.0. (1.0 = 1 metre).

*rolloffscale*

Scaling factor for 3D sound rolloff or attenuation for [FMOD\\_3D\\_LOGROLLOFF](#) based sounds only (which is the default type). Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The doppler scale is a general scaling factor for how much the pitch varies due to doppler shifting in 3D sound. Doppler is the pitch bending effect when a sound comes towards the listener or moves away from it, much like the effect you hear when a train goes past you with its horn sounding. With dopplerscale you can exaggerate or diminish the effect.

FMOD's effective speed of sound at a doppler factor of 1.0 is 340 m/s.

The distance factor is the FMOD 3D engine relative distance factor, compared to 1.0 meters.

Another way to put it is that it equates to "how many units per meter' does your engine have".

For example. If you are using feet then scale would equal 3.28.

Note! This only affects doppler! If you keep your min/max distance, custom rolloff curves and positions in scale relative to each other the volume rolloff will not change.

If you set this, the mindistance of a sound will automatically set itself to this value when it is created in case the user forgets to set the mindistance to match the new distancefactor.

The rolloff scale sets the global attenuation rolloff factor for [FMOD\\_3D\\_LOGROLLOFF](#) based sounds only (which

is the default).

Normally volume for a sound will scale at mindistance / distance. This gives a logarithmic attenuation of volume as the source gets further away (or closer).

Setting this value makes the sound drop off faster or slower. The higher the value, the faster volume will attenuate, and conversely the lower the value, the slower it will attenuate.

For example a rolloff factor of 1 will simulate the real world, where as a value of 2 will make sounds attenuate 2 times quicker.

rolloffscale has no effect for [FMOD\\_3D\\_LINEARROLLOFF](#) or [FMOD\\_3D\\_CUSTOMROLLOFF](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::get3DSettings](#)
- [\\_Sound::set3DMinMaxDistance](#)
- [\\_Sound::get3DMinMaxDistance](#)
- [\\_Channel::set3DAttributes](#)
- [\\_Channel::get3DAttributes](#)

# System::set3DSpeakerPosition

This function allows the user to specify the position of their actual physical speaker to account for non standard setups.

It also allows the user to disable speakers from 3D consideration in a game.

The function is for describing the 'real world' speaker placement to provide a more natural panning solution for 3d sound. Graphical configuration screens in an application could draw icons for speaker placement that the user could position at their will.

## Syntax

```
FMOD_RESULT System::set3DSpeakerPosition(
    FMOD_SPEAKER speaker,
    float x,
    float y,
    bool active
);
```

## Parameters

*speaker*

The selected speaker of interest to position.

*x*

The 2D X offset in relation to the listening position. For example -1.0 would mean the speaker is on the left, and +1.0 would mean the speaker is on the right. 0.0 is the speaker is in the middle.

*y*

The 2D Y offset in relation to the listening position. For example -1.0 would mean the speaker is behind the listener, and +1 would mean the speaker is in front of the listener.

*active*

Enables or disables speaker from 3D consideration. Useful for disabling center speaker for vocals for example, or the LFE. x and y can be anything in this case.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

**Note!** This only affects software mixed 3d sounds, created with [FMOD\\_SOFTWARE](#) and [FMOD\\_3D](#).

A typical 7.1 setup would look like this.

```

sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_FRNT\_E\_FT, -1.0f, 1.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_FRNT\_RG\_HT, 1.0f, 1.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_FRNT\_CE\_NE\_R, 0.0f, 1.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_QW\_FROUE\_NY, 0.0f, 0.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_BCK\_E\_FT, -1.0f, -1.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_BCK\_RG\_HT, 1.0f, -1.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_RSI\_E\_E\_FT, -1.0f, 0.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_RSI\_E\_RG\_HT, 1.0f, 0.0f, tne);

```

A typical stereo setup would look like this.

```

sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_FRNT\_E\_FT, -1.0f, 0.0f, tne);
sys em- ze t t pake rBsi tio n(FMO\_DS\_EAKE\_R\_FRNT\_RG\_HT, 1.0f, 0.0f, tne);

```

You could use this function to make sounds in front of your come out of different physical speakers. If you specified for example that [FMOD\\_SPEAKER\\_SIDE\\_RIGHT](#) was in front of you at and you organized the other speakers accordingly the 3d audio would come out of the side right speaker when it was in front instead of the default which is only to the side.

This function is also useful if speakers are not 'perfectly symmetrical'. For example if the center speaker was closer to the front left than the front right, this function could be used to position that center speaker accordingly and FMOD would skew the panning appropriately to make it sound correct again.

The 2d coordinates used are only used to generate angle information. Size / distance does not matter in FMOD's implementation because it is not FMOD's job to attenuate or amplify the signal based on speaker distance. If it amplified the signal in the digital domain the audio could clip/become distorted. It is better to use the amplifier's analogue level capabilities to balance speaker volumes.

Calling [System::setSpeakerMode](#) overrides these values, so this function must be called after this.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::get3DSpeakerPosition](#)
- [System::setSpeakerMode](#)
- [FMOD\\_SPEAKERMODE](#)
- [FMOD\\_SPEAKER](#)

# System::setAdvancedSettings

Sets advanced features like configuring memory and cpu usage for [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) usage.?

## Syntax

```
FMOD_RESULT System::setAdvancedSettings (
    FMOD_ADVANCEDSETTINGS * settings
);
```

## Parameters

*settings*

Pointer to [FMOD\\_ADVANCEDSETTINGS](#) structure.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_ADVANCEDSETTINGS](#)
- [System::getAdvancedSettings](#)
- [FMOD\\_MODE](#)

# System::setCallback

Sets a callback for a system for a specific event.?

## Syntax

```
FMOD_RESULT System::setCallback (
    FMOD_SYSTEM_CALLBACK_TYPE type,
    FMOD_SYSTEM_CALLBACK callback
);
```

## Parameters

*type*

The callback type, for example a 'device list changed' callback.

*callback*

Pointer to a callback to receive the event when it happens.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Currently some system callbacks are driven by [System::update](#) and will only occur when this function is called. This has the main advantage of far less complication due to thread issues, and allows all FMOD commands, including loading sounds and playing new sounds from the callback.

The only disadvantage is that callbacks are not asynchronous and are bound by the latency caused by the rate the user calls the update command.

Callbacks are stdcall. Use F\_CALLBACK inbetween your return type and function name.

Example:

```
FMOD_RESULT F_CALLBACK mycallback (FMOD_SYSTEM system, FMOD\_SYSTEM\_CALLBACK\_TYPE type,
unsigned int command1, unsigned int command2)
{

    FMOD_System_Cppsystem = (FMOD_System *)system;

    // More code goes here.

    return FMOD\_OK;
}
```

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System:update](#)
- [\\_FMOD\\_SYSTEM\\_CALLBACK](#)
- [\\_FMOD\\_SYSTEM\\_CALLBACKTYPE](#)

# System::setDSPBufferSize

Sets the FMOD internal mixing buffer size. This function is used if you need to control mixer latency or granularity. Smaller buffersizes lead to smaller latency, but can lead to stuttering/skipping/instable sound on slower machines or soundcards with bad drivers.

## Syntax

```
FMOD_RESULT Sys::setDSPBufferSize(
    unsigned int    bufferlength,
    int             numbuffers
);
```

## Parameters

*bufferlength*

The mixer engine block size in samples. Use this to adjust mixer update granularity. Default = 1024. (milliseconds = 1024 at 48khz = 1024 / 48000 \* 1000 = 21.33ms). This means the mixer updates every 21.33ms.

*numbuffers*

The mixer engine number of buffers used. Use this to adjust mixer latency. Default = 4. To get the total buffersize multiply the bufferlength by the numbuffers value. By default this would be 4\*1024.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The FMOD software mixer mixes to a ringbuffer. The size of this ringbuffer is determined here. It mixes a block of sound data every 'bufferlength' number of samples, and there are 'numbuffers' number of these blocks that make up the entire ringbuffer.

Adjusting these values can lead to extremely low latency performance (smaller values), or greater stability in sound output (larger values).

Warning! The 'buffersize' is generally best left alone. Making the granularity smaller will just increase CPU usage (cache misses and DSP network overhead). Making it larger affects how often you hear commands update such as volume/pitch/pan changes. Anything above 20ms will be noticable and sound parameter changes will be obvious instead of smooth.

FMOD chooses the most optimal size by default for best stability, depending on the output type, and if the drivers are emulated or not (for example DirectSound is emulated using waveOut on NT). It is not recommended changing this value unless you really need to. You may get worse performance than the default settings chosen by FMOD.

To convert from milliseconds to 'samples', simply multiply the value in milliseconds by the sample rate of the output (ie 48000 if that is what it is set to), then divide by 1000.



The values in milliseconds and average latency expected from the settings can be calculated using the following code.

```
FMOD_RESULT result;
unsigned int blocks;
int numblocks;
float ms;

result = system->getDSPBufferSize(&blocks, &numblocks);
result = system->getSoftwareFormat(&format, &channels, &rate, &bits);

ms = (float)blocks * 1000.0f / (float)rate;

printf("Min blocks = %.02fms\n", ms);
printf("Min total buffer size = %.02fms\n", ms * numblocks);
printf("Min average latency = %.02fms\n", ms * ((float)numblocks - 1.5f));
```

**Platform notes:** Some output modes (such as [FMOD\\_OUTPUTTYPE\\_ASIO](#)) will change the buffer size to match their own internal optimal buffer size. Use [System::getDSPBufferSize](#) after calling [System::init](#) to see if this is the case. Linux output modes will ignore numbuffers and just write the buffer size to the output every time it can. It does not use a ringbuffer.

Xbox 360 defaults to 256 sample buffersize and 4 for numblocks. This gives a 5.333ms granularity with roughly a 10-15ms latency.

This function cannot be called after FMOD is already activated with [System::init](#). It must be called before [System::init](#), or after [System::close](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getDSPBufferSize](#)
- [System::getSoftwareFormat](#)
- [System::init](#)
- [System::close](#)

# System::setDriver

Selects a soundcard driver. This function is used when an output mode has enumerated more than one output device, and you need to select between them.

## Syntax

```
FMOD_RESULT System::setDriver(  
    int driver  
);
```

## Parameters

*driver*

Driver number to select. 0 = primary or main sound device as selected by the operating system settings. Use [System::getNumDrivers](#) to select a specific device.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If this function is called after FMOD is already initialized with [System::init](#), the current driver will be shutdown and the newly selected driver will be initialized / started.

When switching output driver after [System::init](#) there are a few considerations to make:

All sounds must be created with [FMOD\\_SOFTWARE](#), creating even one [FMOD\\_HARDWARE](#) sound will cause this function to return [FMOD\\_ERR\\_NEEDSSOFTWARE](#).

The driver that you wish to change to must support the current output format, sample rate, and number of channels. If it does not, [FMOD\\_ERR\\_OUTPUT\\_INIT](#) is returned and driver state is cleared. You should now call [System::setDriver](#) with your original driver index to restore driver state (providing that driver is still available / connected) or make another selection.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getDriver](#)

- [\\_System::getNumDrivers](#)
- [\\_System::getDriverInfo](#)
- [\\_System::setOutput](#)
- [\\_System::init](#)
- [\\_System::close](#)

Version 4.12.03 Built on Feb 18, 2008

# System::setFileSystem

Specify user callbacks for FMOD's internal file manipulation functions.

?If ANY of the callback functions are set to 0/ NULL, then FMOD will switch back to its own file routines.

?This function is useful for replacing FMOD's file system with a game system's own file reading API.

?

## Syntax

```
FMOD_RESULT System::setFileSystem(  
    FMOD_FILE_OPEN_CALLBACK useOpen,  
    FMOD_FILE_CLOSE_CALLBACK useClose,  
    FMOD_FILE_READ_CALLBACK useRead,  
    FMOD_FILE_SEEK_CALLBACK useSeek,  
    int blockalign  
);
```

## Parameters

*useropen*

Callback for opening a file. Specifying 0 / null will disable file callbacks.

*userclose*

Callback for closing a file. Specifying 0 / null will disable file callbacks.

*userread*

Callback for reading from a file. Specifying 0 / null will disable file callbacks.

*userseek*

Callback for seeking within a file. Specifying 0 / null will disable file callbacks.

*blockalign*

Internal minimum file block alignment. FMOD will read data in at least chunks of this size if you ask it to. Specifying 0 means there is no file buffering at all (this could adversely affect streaming). Do NOT make this a large value, it is purely a setting for minimum sector size alignment to aid seeking and reading on certain media. It is not for stream buffer sizes, that is what System::setStreamBufferSize is for. It is recommended just to pass -1. Large values just mean large memory usage with no benefit. Specify -1 to not set this value. Default = 2048.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This has no effect on sounds loaded with [FMOD\\_OPENMEMORY](#) or FMOD\_CREATEUSER.

This function can be used to set user file callbacks, or if required, they can be turned off by specifying 0.

This function can be used purely to set the 'buffersize' parameter, and ignore the callback aspect of the function.

Warning : This function can cause unpredictable behaviour if not used properly. You must return the right values, and each command must work properly, or FMOD will not function, or it may even crash if you give it invalid data.

You must also return [FMOD\\_ERR\\_FILE\\_EOF](#) from a read callback if the number of bytes read is smaller than the number of bytes requested.

FMOD's default filesystem buffers reads every 2048 bytes by default. This means every time fmod reads one byte from the API (say if it was parsing a file format), it simply mem copies the byte from the 2k memory buffer, and every time it needs to, refreshes the 2k buffer resulting in a drastic reduction in file I/O. Large reads go straight to the pointer instead of the 2k buffer if it is buffer aligned. This value can be increased or decreased by the user. A buffer of 0 means all reads go directly to the pointer specified. 2048 bytes is the size of a CD sector on most CD ISO formats so it is chosen as the default, for optimal reading speed from CD media.

**NOTE!** Do not force a cast from your function pointer to the FMOD\_FILE\_XXXCALLBACK type! Never try to 'force' fmod to accept your function. If there is an error then find out what it is. Remember to include F\_CALLBACK between the return type and the function name, this equates to stdcall which you must include otherwise (besides not compiling) it will cause problems such as crashing and callbacks not being called.

**NOTE!** Your file callbacks must be thread safe. If not unexpected behaviour may occur. FMOD calls file functions from asynchronous threads, such as the streaming thread, and thread related to [FMOD\\_NONBLOCKING](#) flag.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::init](#)
- [\\_System::attachFileSystem](#)
- [FMOD\\_FILE\\_OPENCALLBACK](#)
- [FMOD\\_FILE\\_CLOSECALLBACK](#)
- [FMOD\\_FILE\\_READCALLBACK](#)
- [FMOD\\_FILE\\_SEEKCALLBACK](#)

# System::setGeometrySettings

Sets the maximum world size for the geometry engine for performance / precision reasons.?

## Syntax

```
FMOD_RESULT System::setGeometrySettings (
    float maxWorldSize
);
```

## Parameters

*maxWorldSize*

Maximum size of the world from the centerpoint to the edge using the same units used in other 3D functions.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Setting maxWorldSize should be done first before creating any geometry.

It can be done any time afterwards but may be slow in this case.

Objects or polygons outside the range of maxWorldSize will not be handled efficiently.

Conversely, if maxWorldSize is excessively large, the structure may lose precision and efficiency may drop.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createGeometry](#)
- [System::getGeometrySettings](#)

# System::setHardwareChannels

This function allows the user to request a minimum number of hardware voices to be present on the soundcard to allow hardware 3D sound acceleration, or clamp the number of hardware 3D voices to a maximum value.?

## Syntax

```
FMOD_RESULT System::setHardwareChannels (
    int min2d,
    int max2d,
    int min3d,
    int max3d
);
```

## Parameters

*min2d*

Minimum number of hardware voices on a soundcard required to actually support hardware 2D sound. If the soundcard does not match this value for number of hardware voices possible, FMOD will place the sound into software mixed buffers instead hardware mixed buffers to guarantee the number of sounds playable at once is guaranteed.

*max2d*

Maximum number of hardware voices to be used by FMOD. This clamps the polyphony of hardware 2D voices to a user specified number. This could be used to limit the number of 2D hardware voices possible at once so that it doesn't sound noisy, or the user might want to limit the number of channels used for 2D hardware support to avoid problems with certain buggy soundcard drivers that report they have many channels but actually don't.

*min3d*

Minimum number of hardware voices on a soundcard required to actually support hardware 3D sound. If the soundcard does not match this value for number of hardware voices possible, FMOD will place the sound into software mixed buffers instead hardware mixed buffers to guarantee the number of sounds playable at once is guaranteed.

*max3d*

Maximum number of hardware voices to be used by FMOD. This clamps the polyphony of hardware 3D voices to a user specified number. This could be used to limit the number of 3D hardware voices possible at once so that it doesn't sound noisy, or the user might want to limit the number of channels used for 3D hardware support to avoid problems with certain buggy soundcard drivers that report they have many channels but actually don't.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 'min' value sets the minimum allowable hardware channels before FMOD drops back to 100 percent software based buffers for sounds even if they are allocated with [FMOD\\_HARDWARE](#). This is helpful for minimum spec cards, and not having to 'guess' how many hardware channels they might have. This way you can guarantee and assume a certain number of channels for your application and always allocate with [FMOD\\_HARDWARE](#) | [FMOD\\_3D](#) without fear of the playsound failing.

The 'max' value function has nothing to do with the 'min' value, in that this is not a function that forces FMOD channels into software mode if a card has less than or more than a certain number of channels.

This parameter only sets a limit on hardware channels playable at once, so if your card has 96 hardware channels, and you set max to 10, then you will only have 10 hardware 3D channels to use.

The 'buggy soundcard driver' issue in the description for the 'max' parameter is to do with one known sound card driver in particular, the default Windows XP SoundBlaster Live drivers. They report over 32 possible voices, but actually only support 32, and when you use the extra voices the driver can act unpredictably causing either sound dropouts or a crash.

This function cannot be called after FMOD is already activated with [System::init](#).

It must be called before [System::init](#), or after [System::close](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getHardwareChannels](#)
- [System::init](#)
- [System::close](#)



# System::setNetworkProxy

Set a proxy server to use for all subsequent internet connections.?

## Syntax

```
FMOD_RESULT Sys::setNetworkProxy (
    const char * proxy
);
```

## Parameters

*proxy*

The name of a proxy server in host:port format e.g. www.fmod.org:8888 (defaults to port 80 if no port is specified).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Basic authentication is supported. To use it, this parameter must be in user:password@host:port format e.g. bob:sekrit123@www.fmod.org:8888 Set this parameter to 0 / NULL if no proxy is required.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, PlayStation 3

## See Also

- [System::getNetworkProxy](#)

# System::setNetworkTimeout

Set the timeout for network streams.?

## Syntax

```
FMOD_RESULT System::setNetworkTimeout(  
    int timeout  
);
```

## Parameters

*timeout*

The timeout value in ms.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, PlayStation 3

## See Also

- [System::getNetworkTimeout](#)

# System::setOutput

This function selects the output mode for the platform. This is for selecting different OS specific API's which might have different features.?

## Syntax

```
FMOD_RESULT System::setOutput(  
    FMOD_OUTPUTTYPE output  
);
```

## Parameters

*output*

Output type to select. See type list for different output types you can select.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is not necessary to call. It is only if you want to specifically switch away from the default output mode for the operating system. The most optimal mode is selected by default for the operating system. For example **FMOD\_OUTPUTTYPE\_DSOUND** is selected on all operating systems except for Windows NT, where [FMOD\\_OUTPUTTYPE\\_WINMM](#) is selected because it is lower latency / faster.

This function cannot be called after FMOD is already activated with [System::init](#).  
It must be called before [System::init](#), or after [System::close](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_OUTPUTTYPE](#)
- [System::init](#)
- [System::close](#)



# System::setOutputByPlugin

Selects an output type based on the enumerated list of outputs including FMOD and 3rd party output plugins.?

## Syntax

```
FMOD_RESULT System::setOutputByPlugin(  
    int index  
);
```

## Parameters

*index*

Index into the enumerated list of output plugins.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function cannot be called after FMOD is already activated with [System::init](#).

It must be called before [System::init](#), or after [System::close](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getNumPlugins](#)
- [System::getOutputByPlugin](#)
- [System::setOutput](#)
- [System::init](#)
- [System::close](#)

# System::setPluginPath

Specify a base search path for plugins so they can be placed somewhere else than the directory of the main executable.

## Syntax

```
FMOD_RESULT System::setPluginPath(
    const char * path
);
```

## Parameters

*path*

A character string containing a correctly formatted path to load plugins from.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 'plugin' version of FMOD relies on plugins, so when [System::init](#) is called it tries to load all FMOD registered plugins.

This path is where it will attempt to load from.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::loadPlugin](#)
- [System::init](#)

# System::setRecordDriver

Selects a recording driver.

?This function is used when an output mode has enumerated more than one record device, and you need to select between them.?

## Syntax

```
FMOD_RESULT System::setRecordDriver(  
    int driver  
);
```

## Parameters

*driver*

Record driver number to select.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function must be called before [System::recordStart](#) or after System::recordStop.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Solaris

## See Also

- [System::getRecordDriver](#)
- [System::getRecordNumDrivers](#)
- [System::GetRecordDriverInfo](#)
- [System::recordStart](#)
- [System::setOutput](#)
- [System::init](#)
- [System::close](#)





# System::setReverbAmbientProperties

Sets a 'background' default reverb environment for the virtual reverb system. This is a reverb preset that will be morphed to if the listener is not within any virtual reverb zones.

?By default the ambient reverb is set to 'off'.?

## Syntax

```
FMOD_RESULT System::setReverbAmbientProperties (
    FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of a [FMOD\\_REVERB\\_PROPERTIES](#) structure containing the settings for the desired ambient reverb setting.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

There is one reverb DSP dedicated to providing a 3D reverb effect. This DSP's properties are a weighted sum of all the contributing virtual reverbs.

The default 3d reverb properties specify the reverb properties in the 3D volumes which has no virtual reverbs defined.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [System::getReverbAmbientProperties](#)
- [System::createReverb](#)



# System::setReverbProperties

Sets parameters for the global reverb environment.

?Reverb parameters can be set manually, or automatically using the pre-defined presets given in the fmod.h header.?

## Syntax

```
FMOD_RESULT System::setReverbProperties (
    const FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of an [FMOD\\_REVERB\\_PROPERTIES](#) structure which defines the attributes for the reverb.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

With [FMOD\\_HARDWARE](#) on Windows using EAX, the reverb will only work on [FMOD\\_3D](#) based sounds. [FMOD\\_SOFTWARE](#) does not have this problem and works on [FMOD\\_2D](#) and [FMOD\\_3D](#) based sounds.

On PlayStation 2, the reverb is limited to only a few reverb types that are not configurable. Use the FMOD\_PRESET\_PS2\_xxx presets.

On Xbox, it is possible to apply reverb to [FMOD\\_2D](#) and [FMOD\\_HARDWARE](#) based voices using this function. By default reverb is turned off for [FMOD\\_2D](#) hardware based voices.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [System::getReverbProperties](#)
- [Channel::setReverbProperties](#)
- [Channel::getReverbProperties](#)



# System::setSoftwareChannels

Sets the maximum number of software mixed channels possible. Software mixed voices are used by sounds loaded with [FMOD\\_SOFTWARE](#).

## Syntax

```
FMOD_RESULT System::setSoftwareChannels (
    int numsoftwarechannels
);
```

## Parameters

*numsoftwarechannels*

The maximum number of [FMOD\\_SOFTWARE](#) mixable voices to be allocated by FMOD. If you don't require software mixed voices specify 0. Default = 32.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

32 voices are allocated by default to be played simultaneously in software.

To turn off the software mixer completely including hardware resources used for the software mixer, specify [FMOD\\_INIT\\_SOFTWARE\\_DISABLE](#) in [System::init](#).

This function cannot be called after FMOD is already activated with [System::init](#).

It must be called before [System::init](#), or after [System::close](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MODE](#)
- [FMOD\\_INITFLAGS](#)
- [System::init](#)
- [System::close](#)
- [System::getSoftwareChannels](#)



# System::setSoftwareFormat

Sets the output format for the software mixer. This includes the bitdepth, sample rate and number of output channels. Do not call this unless you explicitly want to change something. Calling this could have adverse impact on the performance and panning behaviour.

## Syntax

```
FMOD_SYSTEM::setSoftwareFormat(  
    int sampleRate,  
    FMOD_SOUND_FORMAT format,  
    int numOutputChannels,  
    int maxInputChannels,  
    FMOD_DSP_RESAMPLER resampleMethod  
);
```

## Parameters

*sampleRate*

The soundcard's output rate. default = 48000.

*format*

The soundcard's output format. default = [FMOD\\_SOUND\\_FORMAT\\_PCM16](#).

*numoutputchannels*

The number of output channels / speakers to initialize the soundcard to. 0 = keep speaker mode setting (set with [System::setSpeakerMode](#)). If anything else than 0 is specified then the speaker mode will be overridden and will become [FMOD\\_SPEAKERMODE\\_RAW](#), meaning logical speaker assignments (as defined in [FMOD\\_SPEAKER](#)) become ineffective and cannot be used. [Channel::setPan](#) will also fail. Default = 2 ([FMOD\\_SPEAKERMODE\\_STEREO](#)).

*maxinputchannels*

Optional. Specify 0 to ignore. Default = 6. Maximum channel count in loaded/created sounds to be supported. This is here purely for memory considerations and affects how much memory is used in the software mixer when allocating matrices for panning. Do not confuse this with recording, or anything to do with how many voices you can play at once. This is purely for setting the largest type of sound you can play (ie 1 = mono, 2 = stereo, etc.). Most of the time the user will not play sounds any larger than mono or stereo, so setting this to 2 would save memory and cover most sounds that are playable.

*resamplemethod*

Software engine resampling method. default = [FMOD\\_DSP\\_RESAMPLER\\_LINEAR](#). See [FMOD\\_DSP\\_RESAMPLER](#) for different types.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

**Note!** The settings in this function *may* be overridden by the output mode.

FMOD\_OUTPUTTYPE\_ASIO will always change the output mode to [FMOD\\_SOUND\\_FORMAT\\_PCMFLOAT](#) to be compatible with the output formats selectable by the ASIO control panel.

FMOD\_OUTPUTTYPE\_ASIO will also change the samplerate specified by the user to the one selected in the ASIO control panel.

Use [System::getSoftwareFormat](#) after [System::init](#) to determine what the output has possibly changed the format to. Call it after [System::init](#).

It is dependant on the output whether it will force a format change and override these settings or not.

If the output does not support the output mode specified [System::init](#) will fail, and you will have to try another setting.

**Note!** When this function is called with a output channel count greater than 0, the speaker mode is set to [FMOD\\_SPEAKERMODE\\_RAW](#). FMOD does not know when you specify a number of output channels what type of speaker system it is connected to, so [Channel::setPan](#) or [Channel::setSpeakerMix](#) will then fail to work. Calling [System::setSpeakerMode](#) will override the output channel speaker count.

This function cannot be called after FMOD is already activated with [System::init](#).

It must be called before [System::init](#), or after [System::close](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getSoftwareFormat](#)
- [System::setSpeakerMode](#)
- [System::init](#)
- [System::close](#)
- [Channel::setPan](#)
- [Channel::setSpeakerMix](#)
- [FMOD\\_SPEAKER](#)
- [FMOD\\_SPEAKERMODE](#)
- [FMOD\\_SOUND\\_FORMAT](#)
- [FMOD\\_DSP\\_RESAMPLER](#)



# System::setSpeakerMode

Sets the speaker mode in the hardware and FMOD software mixing engine.?

## Syntax

```
FMOD_RESULT System::setSpeakerMode(
    FMOD_SPEAKERMODE speakermode
);
```

## Parameters

*speakermode*

Speaker mode specified from the list in [FMOD\\_SPEAKERMODE](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Speaker modes that are supported on each platform are as follows.

Win32 - [FMOD\\_SPEAKERMODE\\_RAW](#), [FMOD\\_SPEAKERMODE\\_MONO](#),  
[FMOD\\_SPEAKERMODE\\_STEREO](#), [FMOD\\_SPEAKERMODE\\_QUAD](#),  
[FMOD\\_SPEAKERMODE\\_SURROUND](#), [FMOD\\_SPEAKERMODE\\_5POINT1](#),  
[FMOD\\_SPEAKERMODE\\_7POINT1](#), [FMOD\\_SPEAKERMODE\\_PROLOGIC](#).  
Win64 - [FMOD\\_SPEAKERMODE\\_RAW](#), [FMOD\\_SPEAKERMODE\\_MONO](#),  
[FMOD\\_SPEAKERMODE\\_STEREO](#), [FMOD\\_SPEAKERMODE\\_QUAD](#),  
[FMOD\\_SPEAKERMODE\\_SURROUND](#), [FMOD\\_SPEAKERMODE\\_5POINT1](#),  
[FMOD\\_SPEAKERMODE\\_7POINT1](#), [FMOD\\_SPEAKERMODE\\_PROLOGIC](#).  
Linux - [FMOD\\_SPEAKERMODE\\_RAW](#), [FMOD\\_SPEAKERMODE\\_MONO](#),  
[FMOD\\_SPEAKERMODE\\_STEREO](#), [FMOD\\_SPEAKERMODE\\_PROLOGIC](#).  
Mac - [FMOD\\_SPEAKERMODE\\_RAW](#), [FMOD\\_SPEAKERMODE\\_MONO](#),  
[FMOD\\_SPEAKERMODE\\_STEREO](#), [FMOD\\_SPEAKERMODE\\_PROLOGIC](#).  
Xbox - [FMOD\\_SPEAKERMODE\\_RAW](#), [FMOD\\_SPEAKERMODE\\_MONO](#),  
[FMOD\\_SPEAKERMODE\\_STEREO](#), [FMOD\\_SPEAKERMODE\\_5POINT1](#),  
[FMOD\\_SPEAKERMODE\\_PROLOGIC](#).  
PS2 - [FMOD\\_SPEAKERMODE\\_RAW](#), [FMOD\\_SPEAKERMODE\\_MONO](#),  
[FMOD\\_SPEAKERMODE\\_STEREO](#), [FMOD\\_SPEAKERMODE\\_PROLOGIC](#).  
GC - [FMOD\\_SPEAKERMODE\\_RAW](#), [FMOD\\_SPEAKERMODE\\_MONO](#),  
[FMOD\\_SPEAKERMODE\\_STEREO](#), [FMOD\\_SPEAKERMODE\\_PROLOGIC](#).

NOTE! If [System::setSoftwareFormat](#) is called after this function with a valid output channel count, the speakermode is set to [FMOD\\_SPEAKERMODE\\_RAW](#).

If this function is called after [System::setSoftwareFormat](#), then it will overwrite the channel count specified in that

The channel count that is overwritten for each speaker mode is as follows.

- These channel counts are the channel width of the FMOD DSP system, and affect software mixed sounds (sounds created with [FMOD\\_SOFTWARE](#) flag) only.

**Windows note!** Sound will not behave correctly unless your control panel has set the speaker mode to the correct setup.

If using WinMM output, note that some soundcard drivers do not support multichannel output correctly (ie Creative cards). Other soundcards do.

To set the speaker mode to that of the windows control panel, use [`System:getDriverCaps`](#).

MO DS EAKE MO E s pake mo d;

```
result = system_getDriverCaps(0,0,0,0,? // Gets package and board fault driver.
```

```
result = system-escape Mod (escape mod);
```

**Windows note!** If the speakermode is not actually supported (ie even though the user set the speaker mode to 7.1 in windows, the soundcard might not be able to handle it), you will get [FMOD\\_ERR\\_OUTPUT\\_CREATEBUFFER](#) error under Windows. Change the speaker mode to [FMOD\\_SPEAKERMODE\\_STEREO](#) and re-initialize if this happens.

This function cannot be called after FMOD is already activated with [System::init](#).

## Platforms Supported

## See Also

- System::getSpeakerMode
- FMOD\_SPEAKERMODE

- [\\_System:init](#)
- [\\_System:close](#)
- [\\_System:setSoftwareFormat](#)
- [\\_System:set3DSpeakerPosition](#)
- [\\_System:getDriverCaps](#)
- [FMOD\\_RESULT](#)

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# System::setStreamBufferSize

Sets the internal buffersize for streams opened after this call.

?Larger values will consume more memory (see remarks), whereas smaller values may cause buffer under-run/starvation/stuttering caused by large delays in disk access (ie CDROM or netstream), or cpu usage in slow machines, or by trying to play too many streams at once.

?

## Syntax

```
FMOD_RESULT System::setStreamBufferSize(
    unsigned int filebuffersize,
    FMOD_TIMEUNIT filebuffersizetype
);
```

## Parameters

*filebuffersize*

Size of stream file buffer. Default is 16384 ([FMOD\\_TIMEUNIT\\_RAWBYTES](#)).

*filebuffersizetype*

Type of unit for stream file buffer size. Must be [FMOD\\_TIMEUNIT\\_MS](#), [FMOD\\_TIMEUNIT\\_PCM](#), [FMOD\\_TIMEUNIT\\_PCMBYTES](#) or [FMOD\\_TIMEUNIT\\_RAWBYTES](#). Default is [FMOD\\_TIMEUNIT\\_RAWBYTES](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note this function does not affect streams created with [FMOD\\_OPENUSER](#), as the buffer size is specified in [System::createSound](#).

This function does not affect latency of playback. All streams are pre-buffered (unless opened with [FMOD\\_OPENONLY](#)), so they will always start immediately.

Seek and Play operations can sometimes cause a reflush of this buffer.

If [FMOD\\_TIMEUNIT\\_RAWBYTES](#) is used, the memory allocated is 2 \* the size passed in, because fmod allocates a double buffer.

If [FMOD\\_TIMEUNIT\\_MS](#), [FMOD\\_TIMEUNIT\\_PCM](#) or [FMOD\\_TIMEUNIT\\_PCMBYTES](#) is used, and the stream is infinite (such as a shoutcast netstream), then FMOD cannot calculate a compression ratio to work with when the file is opened. This means it will then base the buffersize on [FMOD\\_TIMEUNIT\\_PCMBYTES](#), or in other words the number of PCM bytes, but this will be incorrect for compressed formats.

Use [FMOD\\_TIMEUNIT\\_RAWBYTES](#) for these type (infinite / undetermined length) of streams for more accurate read sizes.

Note to determine the actual memory usage of a stream, including sound buffer and other overhead, use [Memory\\_GetStats](#) before and after creating a sound.

Note that the stream may still stutter if the codec uses a large amount of cpu time, which impacts the smaller, internal 'decode' buffer.

The decode buffer size is changeable via [FMOD\\_CREATESOUNDEXINFO](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_TIMEUNIT](#)
- [System::createSound](#)
- [System::getStreamBufferSize](#)
- [Sound::getOpenState](#)
- [Channel::setMute](#)
- [Memory\\_GetStats](#)
- [FMOD\\_CREATESOUNDEXINFO](#)

# System::setUserData

Sets a user value that the System object will store internally. Can be retrieved with [System::getUserData](#)?

## Syntax

```
FMOD_RESULT System::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the System object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [System::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getUserData](#)

# System::unloadPlugin

Unloads a plugin from memory.?

## Syntax

```
FMOD_RESULT System::unloadPlugin(
    FMOD_PLUGINTYPE pluginType,
    int index
);
```

## Parameters

*pluginType*

Specify the type of plugin type such as [FMOD\\_PLUGINTYPE\\_OUTPUT](#), [FMOD\\_PLUGINTYPE\\_CODEC](#) or [FMOD\\_PLUGINTYPE\\_DSP](#).

*index*

Index into the enumerated list of output plugins.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::getNumPlugins](#)

# System::unlockDSP

Mutual exclusion function to unlock the FMOD DSP engine (which runs asynchronously in another thread) and let it continue executing.?

## Syntax

```
FMOD_RESULT Sys::unlockDSP();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The DSP engine must be locked with [System::lockDSP](#) before this function is called.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::lockDSP](#)



# System::update

Updates the FMOD system. This should be called once per 'game' tick, or once per frame in your application.?

## Syntax

```
FMOD_RESULT System::update();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This updates the following things.

- 3D Sound. 3D positioning will not update if this function is not called.
- Virtual voices. If more voices are played than there are real hardware/software voices, this function must be called to handle the virtualization.
- \* \_NRT output modes. This function must be called to drive the output for these output modes.
- [FMOD\\_INIT\\_STREAM\\_FROM\\_UPDATE](#). This function must be called to update the streamer if this flag has been used.
- Callbacks. This function must be called to fire callbacks if they are specified.
- [FMOD\\_NONBLOCKING](#). This function must be called to make sounds opened with [FMOD\\_NONBLOCKING](#) flag to work properly.

If [FMOD\\_OUTPUTTYPE\\_NOSOUND\\_NRT](#) or [FMOD\\_OUTPUTTYPE\\_WAVWRITER\\_NRT](#) output modes are used, this function also drives the software / DSP engine, instead of it running asynchronously in a thread as is the default behaviour.

This can be used for faster than realtime updates to the decoding or DSP engine which might be useful if the output is the wav writer for example.

If [FMOD\\_INIT\\_STREAM\\_FROM\\_UPDATE](#) is used, this function will update the stream engine. Combining this with the non realtime output will mean smoother captured output.

**Warning!** Do not be tempted to call this function from a different thread to other FMOD commands! This is dangerous and will cause corruption/crashes. This function is not thread safe, and should be called from the same thread as the rest of the FMOD commands.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::init](#)
- [EMOD\\_INITFLAGS](#)
- [EMOD\\_OUTPUTTYPE](#)
- [EMOD\\_MODE](#)

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# Sound Interface

[Sound::addSyncPoint](#)  
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[Sound::set3DMinMaxDistance](#)  
[Sound::setDefaults](#)  
[Sound::setLoopCount](#)  
[Sound::setLoopPoints](#)  
[Sound::setMode](#)  
[Sound::setSoundGroup](#)  
[Sound::setSubSound](#)  
[Sound::setSubSoundSentence](#)  
[Sound::setUserData](#)  
[Sound::setVariations](#)  
[Sound::unlock](#)

# Sound::addSyncPoint

Adds a sync point at a specific time within the sound. These points can be user generated or can come from a wav file with embedded markers.?

## Syntax

```
FMOD_RESULT Sound::addSyncPoint(
    unsigned int offset,
    FMOD_TIMEUNIT offsettype,
    const char * name,
    FMOD_SYNC_POINT ** point
);
```

## Parameters

*offset*

*offsettype*

*name*

*point*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

In sound forge, a marker can be added a wave file by clicking on the timeline / ruler, and right clicking then selecting 'Insert Marker/Region'.

Riff wrapped mp3 files are also supported.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Sound::getNumSyncPoints](#)
- [\\_Sound::getSyncPoint](#)
- [\\_Sound::getSyncPointInfo](#)
- [\\_Sound::deleteSyncPoint](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::deleteSyncPoint

Deletes a syncpoint within the sound. These points can be user generated or can come from a wav file with embedded markers.?

## Syntax

```
FMOD_RESULT Sound::deleteSyncPoint(
    FMOD_SYNCPOINT * point
);
```

## Parameters

*point*

Address of an FMOD\_SYNCPOINT object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

In sound forge, a marker can be added a wave file by clicking on the timeline / ruler, and right clicking then selecting 'Insert Marker/Region'.

Riff wrapped mp3 files are also supported.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getNumSyncPoints](#)
- [Sound::getSyncPoint](#)
- [Sound::getSyncPointInfo](#)
- [Sound::addSyncPoint](#)

# Sound::get3DConeSettings

Retrieves the inside and outside angles of the sound projection cone.?

## Syntax

```
FMOD_RESULT Sound::get3DConeSettings (
    float * insideconeangle ,
    float * outsideconeangle ,
    float * outsidevolume
);
```

## Parameters

*insideconeangle*

Address of a variable that receives the inside angle of the sound projection cone, in degrees. This is the angle within which the sound is at its normal volume. Optional. Specify 0 or NULL to ignore.

*outsideconeangle*

Address of a variable that receives the outside angle of the sound projection cone, in degrees. This is the angle outside of which the sound is at its outside volume. Optional. Specify 0 or NULL to ignore.

*outsidevolume*

Address of a variable that receives the cone outside volume for this sound. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::set3DConeSettings](#)
- [Channel::set3DConeSettings](#)

# Sound::get3DCustomRolloff

Retrieves a pointer to the sound's current custom rolloff curve.?

## Syntax

```
FMOD_RESULT Sound::get3DCustomRolloff(  
    FMOD_VECTOR ** points,  
    int * numpoints  
);
```

## Parameters

*points*

Address of a variable to receive the pointer to the current custom rolloff point list. Optional. Specify 0 or NULL to ignore.

*numpoints*

Address of a variable to receive the number of points in the current custom rolloff point list. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_VECTOR](#)
- [Sound::set3DCustomRolloff](#)
- [Channel::set3DCustomRolloff](#)
- [Channel::get3DCustomRolloff](#)



# Sound::get3DMinMaxDistance

Retrieve the minimum and maximum audible distance for a sound.?

## Syntax

```
FMOD_RESULT Sound::get3DMinMaxDistance (
    float * min,
    float * max
);
```

## Parameters

*min*

Pointer to value to be filled with the minimum volume distance for the sound. See remarks for more on units. Optional. Specify 0 or NULL to ignore.

*max*

Pointer to value to be filled with the maximum volume distance for the sound. See remarks for more on units. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A 'distance unit' is specified by [System::set3DSettings](#). By default this is set to meters which is a distance scale of 1.0. See [System::set3DSettings](#) for more on this.

The default units for minimum and maximum distances are 1.0 and 10,000.0f.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::set3DMinMaxDistance](#)
- [Channel::set3DMinMaxDistance](#)
- [Channel::get3DMinMaxDistance](#)
- [System::set3DSettings](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::getDefaults

Retrieves a sound's default attributes for when it is played on a channel with [System::playSound](#)?

## Syntax

```
FMOD_RESULT Sound::getDefaults (
    float * frequency,
    float * volume,
    float * pan,
    int * priority
);
```

## Parameters

*frequency*

Address of a variable that receives the default frequency for the sound. Optional. Specify 0 or NULL to ignore.

*volume*

Address of a variable that receives the default volume for the sound. Result will be from 0.0 to 1.0. 0.0 = Silent, 1.0 = full volume. Default = 1.0. Optional. Specify 0 or NULL to ignore.

*pan*

Address of a variable that receives the default pan for the sound. Result will be from -1.0 to +1.0. -1.0 = Full left, 0.0 = center, 1.0 = full right. Default = 0.0. Optional. Specify 0 or NULL to ignore.

*priority*

Address of a variable that receives the default priority for the sound when played on a channel. Result will be from 0 to 256. 0 = most important, 256 = least important. Default = 128. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::setDefaults](#)
- [System::createSound](#)

- [\\_System:playSound](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::getFormat

Returns format information about the sound.?

## Syntax

```
FMOD_RESULT Sound::getFormat(
    FMOD_SOUND_TYPE * type,
    FMOD_SOUND_FORMAT * format,
    int * channels,
    int * bits
);
```

## Parameters

*type*

Address of a variable that receives the type of sound. Optional. Specify 0 or NULL to ignore.

*format*

Address of a variable that receives the format of the sound. Optional. Specify 0 or NULL to ignore.

*channels*

Address of a variable that receives the number of channels for the sound. Optional. Specify 0 or NULL to ignore.

*bits*

Address of a variable that receives the number of bits per sample for the sound. This corresponds to [FMOD\\_SOUND\\_FORMAT](#) but is provided as an integer format for convenience. Hardware compressed formats such as VAG, XADPCM, GCADPCM that stay compressed in memory will return 0. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_SOUND\\_TYPE](#)
- [FMOD\\_SOUND\\_FORMAT](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::getLength

Retrieves the length of the sound using the specified time unit.?

## Syntax

```
FMOD_RESULT Sound::getLength(
    unsigned int * length,
    FMOD_TIMEUNIT lengthtype
);
```

## Parameters

*length*

Address of a variable that receives the length of the sound.

*lengthtype*

Time unit retrieve into the length parameter. See [FMOD\\_TIMEUNIT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Certain timeunits do not work depending on the file format. For example [FMOD\\_TIMEUNIT\\_MODORDER](#) will not work with an mp3 file.

A length of 0xFFFFFFFF usually means it is of unlimited length, such as an internet radio stream or MOD/S3M/XM/IT file which may loop forever.

**Warning!** Using a VBR source that does not have an associated length information in milliseconds or pcm samples (such as MP3 or MOD/S3M/XM/IT) may return inaccurate lengths specify [FMOD\\_TIMEUNIT\\_MS](#) or [FMOD\\_TIMEUNIT\\_PCM](#).

If you want FMOD to retrieve an accurate length it will have to pre-scan the file first in this case. You will have to specify [FMOD\\_ACCURATETIME](#) when loading or opening the sound. This means there is a slight delay as FMOD scans the whole file when loading the sound to find the right length in milliseconds or pcm samples, and this also creates a seek table as it does this for seeking purposes.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

**See Also**

- [EMOD\\_TIMEUNIT](#)

Version 4.12.03 Built on Feb 18, 2008



# Sound::getLoopCount

Retrieves the current loop count value for the specified sound.?

## Syntax

```
FMOD_RESULT Sound::getLoopCount(  
    int * loopcount  
);
```

## Parameters

*loopcount*

Address of a variable that receives the number of times a sound will loop by default before stopping. 0 = oneshot. 1 = loop once then stop. -1 = loop forever. Default = -1

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Unlike the channel loop count function, this function simply returns the value set with [Sound::setLoopCount](#). It does not decrement as it plays (especially seeing as one sound can be played multiple times).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Sound::setLoopCount](#)

# Sound::getLoopPoints

Retrieves the loop points for a sound.?

## Syntax

```
FMOD_RESULT Sound::getLoopPoints (
    unsigned int * loopstart,
    FMOD_TIMEUNIT loopstarttype,
    unsigned int * loopend,
    FMOD_TIMEUNIT loopendtype
);
```

## Parameters

*loopstart*

Address of a variable to receive the loop start point. This point in time is played, so it is inclusive. Optional. Specify 0 or NULL to ignore.

*loopstarttype*

The time format used for the returned loop start point. See [FMOD\\_TIMEUNIT](#).

*loopend*

Address of a variable to receive the loop end point. This point in time is played, so it is inclusive. Optional. Specify 0 or NULL to ignore.

*loopendtype*

The time format used for the returned loop end point. See [FMOD\\_TIMEUNIT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_TIMEUNIT](#)
- [Sound::setLoopPoints](#)



# Sound::getMode

Retrieves the mode bits set by the codec and the user when opening the sound.?

## Syntax

```
FMOD_RESULT Sound::getMode (
    FMOD_MODE * mode
);
```

## Parameters

*mode*

Address of a variable that receives the current mode for this sound.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::setMode](#)
- [System::createSound](#)
- [Channel::setMode](#)
- [Channel::getMode](#)

# Sound::getName

Retrieves the name of a sound.?

## Syntax

```
FMOD_RESULT Sound::getName (
    char * name,
    int nameLen
);
```

## Parameters

*name*

Address of a variable that receives the name of the sound.

*namelen*

Length in bytes of the target buffer to receive the string.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

if [FMOD\\_LOWMEM](#) has been specified in [System::createSound](#), this function will return "(null)".

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createSound](#)
- [FMOD\\_MODE](#)

# Sound::getNumSubSounds

Retrieves the number of subsounds stored within a sound.?

## Syntax

```
FMOD_RESULT Sound::getNumSubSounds (
    int * numsubsounds
);
```

## Parameters

*numsubsounds*

Address of a variable that receives the number of subsounds stored within this sound.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A format that has subsounds is usually a container format, such as FSB, DLS, MOD, S3M, XM, IT.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getSubSound](#)

# Sound::getNumSyncPoints

Retrieves the number of sync points stored within a sound. These points can be user generated or can come from a wav file with embedded markers.?

## Syntax

```
FMOD_RESULT Sound::getNumSyncPoints (
    int *    numsyncpoints
);
```

## Parameters

*numsyncpoints*

Address of a variable to receive the number of sync points within this sound.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

In sound forge, a marker can be added a wave file by clicking on the timeline / ruler, and right clicking then selecting 'Insert Marker/Region'.

Riff wrapped mp3 files are also supported.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getSyncPoint](#)
- [Sound::getSyncPointInfo](#)
- [Sound::addSyncPoint](#)
- [Sound::deleteSyncPoint](#)

# Sound::getNumTags

Retrieves the number of tags belonging to a sound.?

## Syntax

```
FMOD_RESULT Sound::getNumTags (
    i nt *   numTags ,
    i nt *   numTagsUpdated
);
```

## Parameters

*numTags*

Address of a variable that receives the number of tags in the sound. Optional. Specify 0 or NULL to ignore.

*numTagsUpdated*

Address of a variable that receives the number of tags updated since this function was last called. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 'numTagsUpdated' parameter can be used to check if any tags have been updated since last calling this function. This can be useful to update tag fields, for example from internet based streams, such as shoutcast or icecast where the name of the song might change.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getTag](#)





# Sound::getOpenState

Retrieves the state a sound is in after [FMOD\\_NONBLOCKING](#) has been used to open it, or the state of the streaming buffer.?

## Syntax

```
FMOD_RESULT Sound::getOpenState (
    FMOD_OPEN_STATE * openstate,
    unsigned int * percentbuffered,
    bool * starving
);
```

## Parameters

*openstate*

Address of a variable that receives the open state of a sound. Optional. Specify 0 or NULL to ignore.

*percentbuffered*

Address of a variable that receives the percentage of the file buffer filled progress of a stream. Optional. Specify 0 or NULL to ignore.

*starving*

Address of a variable that receives the starving state of a sound. If a stream has decoded more than the stream file buffer has ready for it, it will return TRUE. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

**Note:** The return value will be the result of the asynchronous sound create. Use this to determine what happened if a sound failed to open.

**Note:** Always check 'openstate' to determine the state of the sound. Do not assume that if this function returns [FMOD\\_OK](#) then the sound has finished loading.

## Remarks

When a sound is opened with [FMOD\\_NONBLOCKING](#), it is opened and prepared in the background, or asynchronously.

This allows the main application to execute without stalling on audio loads.

This function will describe the state of the asynchronous load routine i.e. whether it has succeeded, failed or is still in progress.

If 'starving' is true, then you will most likely hear a stuttering/repeating sound as the decode buffer loops on itself and replays old data.

Now that this variable exists, you can detect buffer underrun and use something like [Channel::setMute](#) to keep it quiet

until it is not starving any more.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_OPENSTATE](#)
- [FMOD\\_MODE](#)
- [Channel::setMute](#)

# Sound::getSoundGroup

Retrieves the sound's current soundgroup.?

## Syntax

```
FMOD_RESULT Sound::getSoundGroup(
    FMOD_SOUNDGROUP ** soundgroup
);
```

## Parameters

*soundgroup*

Address of a pointer to a SoundGroup to receive the sound's current soundgroup.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

By default a sound is located in the 'master sound group'. This can be retrieved with [System::getMasterSoundGroup](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::setSoundGroup](#)
- [System::getMasterSoundGroup](#)

# Sound::getSubSound

Retrieves a handle to a Sound object that is contained within the parent sound.?

## Syntax

```
FMOD_RESULT Sound::getSubSound(  
    int index,  
    FMOD_Sound ** subSound  
);
```

## Parameters

*index*

Index of the subsound to retrieve within this sound.

*subsound*

Address of a variable that receives the sound object specified.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If the sound is a stream and [FMOD\\_NONBLOCKING](#) was not used, then this call will perform a blocking seek/flush to the specified subsound.

If [FMOD\\_NONBLOCKING](#) was used to open this sound and the sound is a stream, FMOD will do a non blocking seek/flush and set the state of the subsound to [FMOD\\_OPENSTATE\\_SEEKING](#).

The sound won't be ready to be used in this case until the state of the sound becomes [FMOD\\_OPENSTATE\\_READY](#) (or [FMOD\\_OPENSTATE\\_ERROR](#)).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getNumSubSounds](#)
- [Sound::setSubSound](#)
- [System::createSound](#)

- [FMOD\\_MODE](#)
- [FMOD\\_OPENSTATE](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::getSyncPoint

Retrieve a handle to a sync point. These points can be user generated or can come from a wav file with embedded markers.?

## Syntax

```
FMOD_RESULT Sound::getSyncPoint(
    int index,
    FMOD_SYNC_POINT ** point
);
```

## Parameters

*index*

Index of the sync point to retrieve. Use [Sound::getNumSyncPoints](#) to determine the number of syncpoints.

*point*

Address of a variable to receive a pointer to a sync point.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

In sound forge, a marker can be added a wave file by clicking on the timeline / ruler, and right clicking then selecting 'Insert Marker/Region'.

Riff wrapped mp3 files are also supported.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getNumSyncPoints](#)
- [Sound::getSyncPointInfo](#)
- [Sound::addSyncPoint](#)
- [Sound::deleteSyncPoint](#)





# Sound::getSyncPointInfo

Retrieves information on an embedded sync point. These points can be user generated or can come from a wav file with embedded markers.?

## Syntax

```
FMOD_RESULT Sound::getSyncPointInfo (
    FMOD_SYNC_POINT * point,
    char * name,
    int nameLen,
    unsigned int * offset,
    FMOD_TIMEUNIT offsetType
);
```

## Parameters

*point*

Pointer to a sync point. Use [Sound::getSyncPoint](#) to retrieve a syncpoint or [Sound::addSyncPoint](#) to create one.

*name*

Address of a variable to receive the name of the syncpoint. Optional. Specify 0 or NULL to ignore.

*namelen*

Size of buffer in bytes for name parameter. FMOD will only copy to this point if the string is bigger than the buffer passed in. Specify 0 to ignore name parameter.

*offset*

Address of a variable to receive the offset of the syncpoint in a format determined by the offsettype parameter. Optional. Specify 0 or NULL to ignore.

*offsettype*

A timeunit parameter to determine a desired format for the offset parameter. For example the offset can be specified as pcm samples, or milliseconds.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

In sound forge, a marker can be added a wave file by clicking on the timeline / ruler, and right clicking then selecting 'Insert Marker/Region'.

Riff wrapped mp3 files are also supported.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_Sound::getNumSyncPoints](#)
- [\\_Sound::getSyncPoint](#)
- [\\_Sound::addSyncPoint](#)
- [\\_Sound::deleteSyncPoint](#)

# Sound::getSystemObject

Retrieves the parent System object that was used to create this object.?

## Syntax

```
FMOD_RESULT Sound::getSystemObject(
    FMOD::System ** system
);
```

## Parameters

*system*

Address of a pointer that receives the System object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::createSound](#)

# Sound::getTag

Retrieves a descriptive tag stored by the sound, to describe things like the song name, author etc.?

## Syntax

```
FMOD_RESULT Sound::getTag (
    const char * name,
    int index,
    FMOD_TAG * tag
);
```

## Parameters

*name*

Optional. Name of a tag to retrieve. Used to specify a particular tag if the user requires it. To get all types of tags leave this parameter as 0 or NULL.

*index*

Index into the tag list. If the name parameter is null, then the index is the index into all tags present, from 0 up to but not including the numtags value returned by [Sound::getNumTags](#).

If name is not null, then index is the index from 0 up to the number of tags with the same name. For example if there were 2 tags with the name "TITLE" then you could use 0 and 1 to reference them.

Specifying an index of -1 returns new or updated tags. This can be used to pull tags out as they are added or updated.

*tag*

Pointer to a tag structure. This will receive

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The number of tags available can be found with [Sound::getNumTags](#). The way to display or retrieve tags can be done in 3 different ways.

All tags can be continuously retrieved by looping from 0 to the numtags value in [Sound::getNumTags](#) - 1. Updated tags will refresh automatically, and the 'updated' member of the [FMOD\\_TAG](#) structure will be set to true if a tag has been updated, due to something like a netstream changing the song name for example.

Tags could also be retrieved by specifying -1 as the index and only updating tags that are returned. If all tags are retrieved and this function is called the function will return an error of [FMOD\\_ERR\\_TAGNOTFOUND](#).

Specific tags can be retrieved by specifying a name parameter. The index can be 0 based or -1 in the same fashion as described previously.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Sound::getNumTags](#)
- [\\_FMOD\\_TAG](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::getUserData

Retrieves the user value that that was set by calling the [Sound::setUserData](#) function.?

## Syntax

```
FMOD_RESULT Sound::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [Sound::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Sound::setUserData](#)

# Sound::getVariations

Retrieves the current playback behaviour variations of a sound.?

## Syntax

```
FMOD_RESULT Sound::getVariations (
    float * frequencyvar,
    float * volumevar,
    float * panvar
);
```

## Parameters

*frequencyvar*

Address of a variable to receive the frequency variation in hz. Frequency will play at its default frequency, plus or minus a random value within this range. Default = 0.0. Specify 0 or NULL to ignore.

*volumevar*

Address of a variable to receive the volume variation. 0.0 to 1.0. Sound will play at its default volume, plus or minus a random value within this range. Default = 0.0. Specify 0 or NULL to ignore.

*panvar*

Address of a variable to receive the pan variation. 0.0 to 2.0. Sound will play at its default pan, plus or minus a random value within this range. Pan is from -1.0 to +1.0 normally so the range can be a maximum of 2.0 in this case. Default = 0. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::setVariations](#)





# Sound::lock

Returns a pointer to the beginning of the sample data for a sound.

?

## Syntax

```
FMOD_RESULT Sound::lock (
    unsigned int offset,
    unsigned int length,
    void** ptr1,
    void** ptr2,
    unsigned int* len1,
    unsigned int* len2
);
```

## Parameters

*offset*

Offset in *bytes* to the position you want to lock in the sample buffer.

*length*

Number of *bytes* you want to lock in the sample buffer.

*ptr1*

Address of a pointer that will point to the first part of the locked data.

*ptr2*

Address of a pointer that will point to the second part of the locked data. This will be null if the data locked hasn't wrapped at the end of the buffer.

*len1*

Length of data in *bytes* that was locked for ptr1

*len2*

Length of data in *bytes* that was locked for ptr2. This will be 0 if the data locked hasn't wrapped at the end of the buffer.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

You must always unlock the data again after you have finished with it, using [Sound::unlock](#).

With this function you get access to the RAW audio data, for example 8, 16, 24 or 32bit PCM data, mono or stereo data, and on consoles, vag, xadpcm or gcadpcm compressed data. You must take this into consideration when processing the data within the pointer.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Sound::unlock](#)
- [\\_System::createSound](#)

# Sound::readData

Reads data from an opened sound to a specified pointer, using the FMOD codec created internally.  
?This can be used for decoding data offline in small pieces (or big pieces), rather than playing and capturing it, or loading the whole file at once and having to lock / unlock the data.?

## Syntax

```
FMOD_RESULT Sound::readData (
    void * buffer,
    unsigned int lenbytes,
    unsigned int * read
);
```

## Parameters

*buffer*

Address of a buffer that receives the decoded data from the sound.

*lenbytes*

Number of bytes to read into the buffer.

*read*

Number of bytes actually read.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If too much data is read, it is possible [FMOD\\_ERR\\_FILE\\_EOF](#) will be returned, meaning it is out of data. The 'read' parameter will reflect this by returning a smaller number of bytes read than was requested.

As a sound already reads the whole file then closes it upon calling [System::createSound](#) (unless [System::createStream](#) or [FMOD\\_CREATESTREAM](#) is used), this function will not work because the file is no longer open.

Note that opening a stream makes it read a chunk of data and this will advance the read cursor. You need to either use [FMOD\\_OPENONLY](#) to stop the stream pre-buffering or call [Sound::seekData](#) to reset the read cursor.

If [FMOD\\_OPENONLY](#) flag is used when opening a sound, it will leave the file handle open, and FMOD will not read any data internally, so the read cursor will be at position 0. This will allow the user to read the data from the start.

As noted previously, if a sound is opened as a stream and this function is called to read some data, then you will 'miss the start' of the sound.

[Channel::setPosition](#) will have the same result. These function will flush the stream buffer and read in a chunk of audio internally. This is why if you want to read from an absolute position you should use [Sound::seekData](#) and not the previously mentioned functions.

Remember if you are calling `readData` and `seekData` on a stream it is up to you to cope with the side effects that may occur. Information functions such as `Channel::getPosition` may give misleading results. Calling [Channel::setPosition](#) will reset and flush the stream, leading to the time values returning to their correct position.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::seekData](#)
- [FMOD\\_MODE](#)
- [Channel::setPosition](#)
- [System::createSound](#)
- [System::createStream](#)

# Sound::release

Frees a sound object.?

## Syntax

```
FMOD_RESULT Sound::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This will free the sound object and everything created under it.

If this is a stream that is playing as a subsound of another parent stream, then if this is the currently playing subsound (be it a normal subsound playback, or as part of a sentence), the whole stream will stop.

Note - This function will block if it was opened with [FMOD\\_NONBLOCKING](#) and hasn't finished opening yet.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::createSound](#)
- [\\_Sound::getSubSound](#)

# Sound::seekData

Seeks a sound for use with data reading. This is not a function to 'seek a sound' for normal use. This is for use in conjunction with [Sound::readData](#).

## Syntax

```
FMOD_RESULT Sound::seekData (
    unsigned int pcm
);
```

## Parameters

*pcm*

Offset to seek to in PCM samples.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note. If a stream is opened and this function is called to read some data, then it will advance the internal file pointer, so data will be skipped if you play the stream. Also calling position / time information functions will lead to misleading results.

A stream can be reset before playing by setting the position of the channel (ie using [Channel::setPosition](#)), which will make it seek, reset and flush the stream buffer. This will make it sound correct again.

Remember if you are calling readData and seekData on a stream it is up to you to cope with the side effects that may occur.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::readData](#)
- [Channel::setPosition](#)



# Sound::set3DConeSettings

Sets the inside and outside angles of the sound projection cone, as well as the volume of the sound outside the outside angle of the sound projection cone.

## Syntax

```
FMOD_RESULT Sound::set3DConeSettings (
    float insideconeangle,
    float outsideconeangle,
    float outsidevolume
);
```

## Parameters

*insideconeangle*

Inside cone angle, in degrees, from 0 to 360. This is the angle within which the sound is at its normal volume. Must not be greater than outsideconeangle. Default = 360.

*outsideconeangle*

Outside cone angle, in degrees, from 0 to 360. This is the angle outside of which the sound is at its outside volume. Must not be less than insideconeangle. Default = 360.

*outsidevolume*

Cone outside volume, from 0 to 1.0. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::get3DConeSettings](#)
- [Channel::set3DConeSettings](#)





# Sound::set3DCustomRolloff

Point a sound to use a custom rolloff curve. Must be used in conjunction with [FMOD\\_3D\\_CUSTOMROLLOFF](#) flag to be activated.?

## Syntax

```
FMOD_RESULT Sound::set3DCustomRolloff(
    FMOD_VECTOR * points,
    int numpoints
);
```

## Parameters

*points*

An array of [FMOD\\_VECTOR](#) structures where x = distance and y = volume from 0.0 to 1.0. z should be set to 0.

*numpoints*

The number of points in the array.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

**Note!** This function does not duplicate the memory for the points internally. The pointer you pass to FMOD must remain valid until there is no more use for it.

Do not free the memory while in use, or use a local variable that goes out of scope while in use.

Points must be sorted by distance! Passing an unsorted list to FMOD will result in an error.

Set the points parameter to 0 or NULL to disable the points. If [FMOD\\_3D\\_CUSTOMROLLOFF](#) is set and the rolloff curve is 0, FMOD will revert to logarithmic curve rolloff.

Min and maxdistance are meaningless when [FMOD\\_3D\\_CUSTOMROLLOFF](#) is used and the values are ignored.

Here is an example of a custom array of points.

```
FMOD\_VECTOR Rcurve[ 3] =
{
    { 0.0f, 1.0f, 0.0f} ,
    { 2.0f, 0.2f, 0.0f} ,
    { 20.0f, 0.0f, 0.0f}
```

} ;

x represents the distance, y represents the volume. z is always 0.

Distances between points are linearly interpolated.

Note that after the highest distance specified, the volume in the last entry is used from that distance onwards.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MODE](#)
- [FMOD\\_VECTOR](#)
- [\\_Sound::get3DCustomRolloff](#)
- [\\_Channel::set3DCustomRolloff](#)
- [\\_Channel::get3DCustomRolloff](#)

# Sound::set3DMinMaxDistance

Sets the minimum and maximum audible distance for a sound.

?

?MinDistance is the minimum distance that the sound emitter will cease to continue growing louder at (as it approaches the listener).

?Within the mindistance it stays at the constant loudest volume possible. Outside of this mindistance it begins to attenuate.

?MaxDistance is the distance a sound stops attenuating at. Beyond this point it will stay at the volume it would be at maxdistance units from the listener and will not attenuate any more.

?MinDistance is useful to give the impression that the sound is loud or soft in 3d space. An example of this is a small quiet object, such as a bumblebee, which you could set a mindistance of to 0.1 for example, which would cause it to attenuate quickly and dissapear when only a few meters away from the listener.

?Another example is a jumbo jet, which you could set to a mindistance of 100.0, which would keep the sound volume at max until the listener was 100 meters away, then it would be hundreds of meters more before it would fade out.

?

?In summary, increase the mindistance of a sound to make it 'louder' in a 3d world, and decrease it to make it 'quieter' in a 3d world.

?Maxdistance is effectively obsolete unless you need the sound to stop fading out at a certain point. Do not adjust this from the default if you dont need to.

?Some people have the confusion that maxdistance is the point the sound will fade out to, this is not the case.

?

## Syntax

```
FMOD_RESULT Sound::set3DMinMaxDistance (
    float min,
    float max
);
```

## Parameters

*min*

The sound's minimum volume distance in "units". See remarks for more on units.

*max*

The sound's maximum volume distance in "units". See remarks for more on units.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A 'distance unit' is specified by [System::set3DSettings](#). By default this is set to meters which is a distance scale of 1.0. See [System::set3DSettings](#) for more on this.

The default units for minimum and maximum distances are 1.0 and 10,000.0f.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_Sound::get3DMinMaxDistance](#)
- [\\_Channel::set3DMinMaxDistance](#)
- [\\_Channel::get3DMinMaxDistance](#)
- [\\_System::set3DSettings](#)

# Sound::setDefault

Sets a sound's default attributes, so when it is played it uses these values without having to specify them later for each channel each time the sound is played.

## Syntax

```
FMOD_RESULT Sound::setDefault(  
    float frequency,  
    float volume,  
    float pan,  
    int priority  
);
```

## Parameters

*frequency*

Default playback frequency for the sound, in hz. (ie 44100hz).

*volume*

Default volume for the sound. 0.0 to 1.0. 0.0 = Silent, 1.0 = full volume. Default = 1.0.

*pan*

Default pan for the sound. -1.0 to +1.0. -1.0 = Full left, 0.0 = center, 1.0 = full right. Default = 0.0.

*priority*

Default priority for the sound when played on a channel. 0 to 256. 0 = most important, 256 = least important. Default = 128.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

There are no 'ignore' values for these parameters. Use [Sound::getDefault](#) if you want to change only 1 and leave others unaltered.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

**See Also**

- [\\_Sound::getDefaults](#)
- [\\_System::playSound](#)
- [\\_System::createSound](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::setLoopCount

Sets a sound, by default, to loop a specified number of times before stopping if its mode is set to [FMOD\\_LOOP\\_NORMAL](#) or [FMOD\\_LOOP\\_BIDI](#)?

## Syntax

```
FMOD_RESULT Sound::setLoopCount(  
    int loopCount  
);
```

## Parameters

*loopcount*

Number of times to loop before stopping. 0 = oneshot. 1 = loop once then stop. -1 = loop forever. Default = -1

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function does not affect [FMOD\\_HARDWARE](#) based sounds that are not streamable. [FMOD\\_SOFTWARE](#) based sounds or any type of sound created with `System::CreateStream` or [FMOD\\_CREATESTREAM](#) will support this function.

Issues with streamed audio. (Sounds created with `System::createStream` or [FMOD\\_CREATESTREAM](#)). When changing the loop count, sounds created with `System::createStream` or [FMOD\\_CREATESTREAM](#) may already have been pre-buffered and executed their loop logic ahead of time, before this call was even made.

This is dependant on the size of the sound versus the size of the stream *decode* buffer. See [FMOD\\_CREATESOUNDEXINFO](#).

If this happens, you may need to reflush the stream buffer. To do this, you can call `Channel::setPosition` which forces a reflush of the stream buffer.

Note this will usually only happen if you have sounds or loop points that are smaller than the stream decode buffer size. Otherwise you will not normally encounter any problems.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getLoopCount](#)



- [System:setStreamBufferSize](#)
- [FMOD\\_CREATEINDEXINFO](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::setLoopPoints

Sets the loop points within a sound.?

## Syntax

```
FMOD_RESULT Sound::setLoopPoints (
    unsigned int loopstart,
    FMOD_TIMEUNIT loopstarttype,
    unsigned int loopend,
    FMOD_TIMEUNIT loopendtype
);
```

## Parameters

*loopstart*

The loop start point. This point in time is played, so it is inclusive.

*loopstarttype*

The time format used for the loop start point. See [FMOD\\_TIMEUNIT](#).

*loopend*

The loop end point. This point in time is played, so it is inclusive.

*loopendtype*

The time format used for the loop end point. See [FMOD\\_TIMEUNIT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Not supported by static sounds created with [FMOD\\_HARDWARE](#).

Supported by sounds created with [FMOD\\_SOFTWARE](#), or sounds of any type (hardware or software) created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#).

If a sound was 1000ms long and you wanted to loop the whole sound, loopstart would be 0, and loopend would be 999,

not 1000.

If loop end is smaller or equal to loop start, it will result in an error.

If loop start or loop end is larger than the length of the sound, it will result in an error.

Issues with streamed audio. (Sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#)).

When changing the loop points, sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#) may

already have been pre-buffered and executed their loop logic ahead of time, before this call was even made.

This is dependant on the size of the sound versus the size of the stream *decode* buffer. See

[FMOD\\_CREATEINDEXINFO](#).

If this happens, you may need to reflush the stream buffer. To do this, you can call [Channel::setPosition](#) which forces a reflush of the stream buffer.

Note this will usually only happen if you have sounds or looppoints that are smaller than the stream decode buffer size. Otherwise you will not normally encounter any problems.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_TIMEUNIT](#)
- [FMOD\\_MODE](#)
- [Sound::getLoopPoints](#)
- [Sound::setLoopCount](#)
- [System::createStream](#)
- [System::setStreamBufferSize](#)
- [Channel::setPosition](#)
- [FMOD\\_CREATEINDEXINFO](#)

# Sound::setMode

Sets or alters the mode of a sound.?

## Syntax

```
FMOD_RESULT Sound::setMode (
    FMOD_MODE mode
);
```

## Parameters

*mode*

Mode bits to set.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When calling this function, note that it will only take effect when the sound is played again with [System::playSound](#). Consider this mode the 'default mode' for when the sound plays, not a mode that will suddenly change all currently playing instances of this sound.

Flags supported:

FMOD\_LOOP\_OFF

FMOD\_LOOP\_NORMAL

FMOD\_LOOP\_BIDI (only works with sounds created with [FMOD\\_SOFTWARE](#). Otherwise it will behave as [FMOD\\_LOOP\\_NORMAL](#))

FMOD\_3D\_HEADRELATIVE

FMOD\_3D\_WORLDRELATIVE

FMOD\_2D (see notes for win32 hardware voices)

FMOD\_3D (see notes for win32 hardware voices)

FMOD\_3D\_LOGROLLOFF

FMOD\_3D\_LINEARROLLOFF

FMOD\_3D\_CUSTOMROLLOFF

FMOD\_3D\_IGNOREGEOMETRY

FMOD\_DONTRESTOREVIRTUAL

Issues with streamed audio. (Sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#)). When changing the loop mode, sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#) may already have been pre-buffered and executed their loop logic ahead of time, before this call was even made.

This is dependant on the size of the sound versus the size of the stream *decode* buffer. See

[FMOD\\_CREATESOUNDEXINFO](#).

If this happens, you may need to reflush the stream buffer. To do this, you can call [Channel::setPosition](#) which forces a reflush of the stream buffer.

Note this will usually only happen if you have sounds or looppoints that are smaller than the stream decode buffer size.

Otherwise you will not normally encounter any problems.

**Win32** [FMOD\\_HARDWARE](#) note. Under DirectSound, you cannot change the mode of a sound between [FMOD\\_2D](#) and [FMOD\\_3D](#). If this is a problem create the sound as [FMOD\\_3D](#) initially, and use [FMOD\\_3D\\_HEADRELATIVE](#) and [FMOD\\_3D\\_WORLDRELATIVE](#). Alternatively just use [FMOD\\_SOFTWARE](#).

If [FMOD\\_3D\\_IGNOREGEOMETRY](#) is not specified, the flag will be cleared if it was specified previously.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MODE](#)
- [Sound::getMode](#)
- [System::setStreamBufferSize](#)
- [System::playSound](#)
- [System::createStream](#)
- [Channel::setPosition](#)
- [FMOD\\_CREATEINDEXINFO](#)

# Sound::setSoundGroup

Moves the sound from its existing SoundGroup to the specified sound group.?

## Syntax

```
FMOD_RESULT Sound::setSoundGroup(
    FMOD_SOUNDGROUP * soundgroup
);
```

## Parameters

*soundgroup*

Address of a SoundGroup object to move the sound to.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

By default a sound is located in the 'master sound group'. This can be retrieved with [System::getMasterSoundGroup](#). Putting a sound in a sound group (or just using the master sound group) allows for functionality like limiting a group of sounds to a certain number of playbacks (see [SoundGroup::setMaxAudible](#)).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getSoundGroup](#)
- [System::getMasterSoundGroup](#)
- [System::createSoundGroup](#)
- [SoundGroup::setMaxAudible](#)

# Sound::setSubSound

Assigns a sound as a 'subsound' of another sound. A sound can contain other sounds. The sound object that is issuing the command will be the 'parent' sound.?

## Syntax

```
FMOD_RESULT Sound::setSubSound(  
    int index,  
    FMOD_Sound * subSound  
);
```

## Parameters

*index*

Index within the sound to set the new sound to as a 'subsound'.

*subsound*

Sound object to set as a subsound within this sound.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getNumSubSounds](#)
- [Sound::getSubSound](#)

# Sound::setSubSoundSentence

For any sound that has subsounds, this function will determine the order of playback of these subsounds, and it will play / stitch together the subsounds without gaps.

?This is a very useful feature for those users wanting to do seamless / gapless stream playback. (ie sports commentary, gapless playback media players etc).?

## Syntax

```
FMOD_RESULT Sound::setSubSoundSentence (
    int * subSoundList,
    int numSubSounds
);
```

## Parameters

*subSoundList*

Pointer to an array of indices which are the subsounds to play. One subsound can be included in this list multiple times if required.

*numSubSounds*

Number of indices inside the subSoundList array.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

**Note!** Only streams can be sentenced. Static samples are not stitchable because most hardware api's don't have a way to gaplessly play 2 sounds after one another.

By default subsounds are stitched automatically from index 0 to the last index. For example a CD that is opened as a sound (and the cd tracks are its subsounds) will play all CD tracks from start to end without gaps if the parent sound is played with [System::playSound](#).

A user can swap subsounds that arent playing at the time to do dynamic stitching/sentencing of sounds.

The currently playing subsound in a sentence can be found with [Channel::getPosition](#) and the timeunit [FMOD\\_TIMEUNIT\\_SENTENCE\\_SUBSOUND](#). This is useful for displaying the currently playing track of a cd in a whole CD sentence for example.

For realtime stitching purposes, it is better to know the buffered ahead of time subsound index. This can be done by adding the flag (using bitwise OR) [FMOD\\_TIMEUNIT\\_BUFFERED](#).

## Platforms Supported



Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

### See Also

- [\\_System::playSound](#)
- [\\_Sound::getSubSound](#)
- [\\_Channel::getPosition](#)
- [\\_FMOD\\_TIMEUNIT](#)

Version 4.12.03 Built on Feb 18, 2008

# Sound::setUserData

Sets a user value that the Sound object will store internally. Can be retrieved with [Sound::getUserData](#)?

## Syntax

```
FMOD_RESULT Sound::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the Sound object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [Sound::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getUserData](#)

# Sound::setVariations

Changes the playback behaviour of a sound by allowing random variations to playback parameters to be set.?

## Syntax

```
FMOD_RESULT Sound::setVariations (
    float frequencyvar,
    float volumevar,
    float panvar
);
```

## Parameters

*frequencyvar*

Frequency variation in hz. Frequency will play at its default frequency, plus or minus a random value within this range. Default = 0.0.

*volumevar*

Volume variation. 0.0 to 1.0. Sound will play at its default volume, plus or minus a random value within this range. Default = 0.0.

*panvar*

Pan variation. 0.0 to 2.0. Sound will play at its default pan, plus or minus a random value within this range. Pan is from -1.0 to +1.0 normally so the range can be a maximum of 2.0 in this case. Default = 0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::getVariations](#)

# Sound::unlock

Releases previous sample data lock from [Sound::lock](#)?

## Syntax

```
FMOD_RESULT Sound::unlock (
    void * ptr1,
    void * ptr2,
    unsigned int len1,
    unsigned int len2
);
```

## Parameters

*ptr1*

Pointer to the 1st locked portion of sample data, from [Sound::lock](#).

*ptr2*

Pointer to the 2nd locked portion of sample data, from [Sound::lock](#).

*len1*

Length of data in *bytes* that was locked for ptr1

*len2*

Length of data in *bytes* that was locked for ptr2. This will be 0 if the data locked hasn't wrapped at the end of the buffer.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

Call this function after data has been read/written to from [Sound::lock](#). This function will do any post processing necessary and if needed, send it to sound ram.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Sound::lock](#)
- [System::createSound](#)

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# Channel Interface

[Channel::addDSP](#)  
[Channel::get3DAttributes](#)  
[Channel::get3DConeOrientation](#)  
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[Channel::getPaused](#)  
[Channel::getPosition](#)  
[Channel::getPriority](#)  
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[Channel::getSpectrum](#)  
[Channel::getSystemObject](#)  
[Channel::getUserData](#)  
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[Channel::set3DConeSettings](#)  
[Channel::set3DCustomRolloff](#)  
[Channel::set3DDopplerLevel](#)  
[Channel::set3DMinMaxDistance](#)  
[Channel::set3DOcclusion](#)  
[Channel::set3DPanLevel](#)  
[Channel::set3DSpread](#)  
[Channel::setCallback](#)  
[Channel::setChannelGroup](#)

[Channel::setDelay](#)  
[Channel::setFrequency](#)  
[Channel::setInputChannelMix](#)  
[Channel::setLoopCount](#)  
[Channel::setLoopPoints](#)  
[Channel::setMode](#)  
[Channel::setMute](#)  
[Channel::setPan](#)  
[Channel::setPaused](#)  
[Channel::setPosition](#)  
[Channel::setPriority](#)  
[Channel::setReverbProperties](#)  
[Channel::setSpeakerLevels](#)  
[Channel::setSpeakerMix](#)  
[Channel::setUserData](#)  
[Channel::setVolume](#)  
[Channel::stop](#)

# Channel::addDSP

This function adds a pre-created DSP unit or effect to the head of the Channel DSP chain.

## Syntax

```
FMOD_RESULT Channel::addDSP(
    FMOD_DSP * dsp
);
```

## Parameters

*dsp*

A pointer to a pre-created DSP unit to be inserted at the head of the Channel DSP chain.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is a wrapper function to insert a DSP unit at the top of the Channel DSP chain. It disconnects the head unit from its input, then inserts the unit at the head and reconnects the previously disconnected input back as an input to the new unit. It is effectively the following code.

```
int numInputs;
Channel::getDSPHead(
    &headGetInputNums);
if (numInputs > 1)
{
    return FMOD_ERR_DSP_TOO_MANY;
}
&headGetInput(0, ?
&head >disconnectFrom(&xt);
&head >addInput(&p);
&p >addInput(&xt);
&p >setTime(tne);
```

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris



## See Also

- [\\_Channel::getDSPHead](#)
- [\\_System::createDSP](#)
- [\\_System::createDSPByType](#)
- [\\_System::createDSPByIndex](#)
- [\\_System::addDSP](#)
- [\\_ChannelGroup::addDSP](#)
- [\\_DSP::remove](#)

# Channel::get3DAttributes

Retrieves the position and velocity of a 3d channel.

## Syntax

```
FMOD_RESULT Channel::get3DAttributes (
    FMOD_VECTOR * pos,
    FMOD_VECTOR * vel
);
```

## Parameters

*pos*

Address of a variable that receives the position in 3D space of the channel. Optional. Specify 0 or NULL to ignore.

*vel*

Address of a variable that receives the velocity in 'distance units per second' in 3D space of the channel. See remarks. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A 'distance unit' is specified by [System::set3DSettings](#). By default this is set to meters which is a distance scale of 1.0.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DAttributes](#)
- [FMOD\\_VECTOR](#)
- [System::set3DSettings](#)



# Channel::get3DConeOrientation

Retrieves the orientation of the sound projection cone for this channel.

## Syntax

```
FMOD_RESULT Channel::get3DConeOrientation(
    FMOD_VECTOR *orientation
);
```

## Parameters

*orientation*

Address of a variable that receives the orientation of the sound projection cone. The vector information represents the center of the sound cone.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DConeOrientation](#)

# Channel::get3DConeSettings

Retrieves the inside and outside angles of the sound projection cone.?

## Syntax

```
FMOD_RESULT Channel::get3DConeSettings (
    float * insideconeangle ,
    float * outsideconeangle ,
    float * outsidevolume
);
```

## Parameters

*insideconeangle*

Address of a variable that receives the inside angle of the sound projection cone, in degrees. This is the angle within which the sound is at its normal volume. Optional. Specify 0 or NULL to ignore.

*outsideconeangle*

Address of a variable that receives the outside angle of the sound projection cone, in degrees. This is the angle outside of which the sound is at its outside volume. Optional. Specify 0 or NULL to ignore.

*outsidevolume*

Address of a variable that receives the cone outside volume for this channel. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DConeSettings](#)
- [Sound::get3DConeSettings](#)

# Channel::get3DCustomRolloff

Retrieves a pointer to the sound's current custom rolloff curve.?

## Syntax

```
FMOD_RESULT Channel::get3DCustomRolloff(
    FMOD_VECTOR ** points,
    int * numpoints
);
```

## Parameters

*points*

Address of a variable to receive the pointer to the current custom rolloff point list. Optional. Specify 0 or NULL to ignore.

*numpoints*

Address of a variable to receive the number of points in the current custom rolloff point list. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_VECTOR](#)
- [Channel::set3DCustomRolloff](#)
- [Sound::set3DCustomRolloff](#)
- [Sound::get3DCustomRolloff](#)

# Channel::get3DDopplerLevel

Retrieves the current 3D doppler level for the channel set by [Channel::set3DDopplerLevel](#)?

## Syntax

```
FMOD_RESULT Channel::get3DDopplerLevel(
    float * pLevel
);
```

## Parameters

*level*

Address of a variable to receives the current doppler scale for this channel. 0 = No doppler. 1 = Normal doppler. 5 = max. Default = 1.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DDopplerLevel](#)

# Channel::get3DMinMaxDistance

Retrieves the current minimum and maximum audible distance for a channel.?

## Syntax

```
FMOD_RESULT Channel::get3DMinMaxDistance (
    float * mindistance,
    float * maxdistance
);
```

## Parameters

*mindistance*

Pointer to a floating point value to store mindistance. Optional. Specify 0 or NULL to ignore.

*maxdistance*

Pointer to a floating point value to store maxdistance. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DMinMaxDistance](#)
- [System::set3DSettings](#)
- [Sound::set3DMinMaxDistance](#)



# Channel::get3DOcclusion

Retrieves the the EAX or software based occlusion factors for a channel.?

## Syntax

```
FMOD_RESULT Channel::get3DOcclusion(  
    float * directocclusion,  
    float * reverbocclusion  
);
```

## Parameters

*directocclusion*

Address of a variable that receives the occlusion factor for a voice for the direct path. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0. Optional. Specify 0 or NULL to ignore.

*reverbocclusion*

Address of a variable that receives the occlusion factor for a voice for the reverb mix. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DOcclusion](#)
- [ChannelGroup::get3DOcclusion](#)

# Channel::get3DPanLevel

Retrieves the current 3D mix level for the channel set by [Channel::set3DPanLevel](#)?

## Syntax

```
FMOD_RESULT Channel::get3DPanLevel(
    float * pLevel
);
```

## Parameters

*level*

0 = Sound pans according to [Channel::setSpeakerMix](#). 1 = Sound pans according to 3d position. Default = 1 (all by 3d position).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DPanLevel](#)
- [Channel::setSpeakerMix](#)

# Channel::get3DSpread

Retrieves the stereo (and above) spread angle specified by [Channel::set3DSpread](#)?

## Syntax

```
FMOD_RESULT Channel::get3DSpread(
    float * angle
);
```

## Parameters

*angle*

Address of a variable that receives the spread angle for subchannels. 0 = all subchannels are located at the same position. 360 = all subchannels are located at the opposite position.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DSpread](#)

# Channel::getAudibility

Returns the combined volume of the channel after 3d sound, volume, channel group volume and geometry occlusion calculations have been performed on it.

## Syntax

```
FMOD_RESULT Channel::getAudibility (  
    float * audibility  
);
```

## Parameters

*audibility*

Address of a variable that receives the channel audibility value.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This does not represent the waveform, just the calculated volume based on 3d distance, occlusion, volume and channel group volume. This value is used by the FMOD Ex virtual channel system to order its channels between real and virtual.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setVolume](#)
- [Channel::getVolume](#)
- [ChannelGroup::setVolume](#)
- [ChannelGroup::getVolume](#)
- [Channel::set3DOcclusion](#)
- [Channel::get3DOcclusion](#)
- [Channel::set3DAttributes](#)
- [Channel::get3DAttributes](#)



# Channel::getChannelGroup

Retrieves the currently assigned channel group for the channel.

## Syntax

```
FMOD_RESULT Channel::getChannelGroup(
    FMOD_CHANNELGROUP** channelgroup
);
```

## Parameters

*channelgroup*

Address of a variable to receive a pointer to the currently assigned channel group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setChannelGroup](#)

# Channel::getCurrentSound

Returns the currently playing sound for this channel.

## Syntax

```
FMOD_RESULT Channel::getCurrentSound(  
    FMOD_SOUND** sound  
);
```

## Parameters

*sound*

Address of a variable that receives the pointer to the currently playing sound for this channel.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a sound is not playing the returned pointer will be 0 or NULL.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::playSound](#)
- [System::playDSP](#)

# Channel::getDSPHead

Returns a pointer to the DSP unit head node that handles software mixing for this channel.  
?Only applicable to channels playing sounds created with [FMOD\\_SOFTWARE](#)?

## Syntax

```
FMOD_RESULT Channel::getDSPHead(
    FMOD_DSP** dsp
);
```

## Parameters

*dsp*

Address of a variable that receives pointer to the current head DSP unit for this channel.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

By default a channel DSP unit usually contains 1 input, which is the wavetable input.

If [System::playDSP](#) has been used then the input to the channel head unit will be the unit that was specified in the call. See the tutorials for more information on DSP networks and how to manipulate them.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createDSP](#)
- [System::createDSPByType](#)
- [System::playDSP](#)



# Channel::getDelay

Sets a delay before the sound is audible and after the sound ends.?

## Syntax

```
FMOD_RESULT Channel::getDelay (
    unsigned int * startDelay ,
    unsigned int * endDelay
);
```

## Parameters

*startDelay*

Address of a variable that receives the current channel delay in milliseconds for before the sound starts. Optional. Specify 0 or NULL to ignore.

*endDelay*

Address of a variable that receives the current channel delay in milliseconds for after the sound stops. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setDelay](#)

# Channel::getFrequency

Returns the frequency in HZ of the channel.

## Syntax

```
FMOD_RESULT Channel::getFrequency (
    float * frequency
);
```

## Parameters

*frequency*

Address of a variable that receives the current frequency of the channel in HZ.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setFrequency](#)

# Channel::getIndex

Retrieves the internal channel index for a channel.?

## Syntax

```
FMOD_RESULTC h na l: ge t i nd e x(  
    i nt * i nd e x  
);
```

## Parameters

*index*

Address of a variable to receive the channel index. This will be from 0 to the value specified in [System::init](#) minus 1.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note that working with channel indices directly is not recommended. It is recommended that you use [FMOD\\_CHANNEL\\_FREE](#) for the index in [System::playSound](#) to use FMOD's channel manager.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::playSound](#)
- [\\_System::init](#)

# Channel::getInputChannelMix

Retrieves the incoming levels for a channel in a sound.

?A mono sound has 1 input channel, a stereo has 2, etc. It depends on what type of sound is playing on the channel at the time.?

## Syntax

```
FMOD_RESULT Channel::getInputChannelMix(
    float * levels,
    int numlevels
);
```

## Parameters

*levels*

Address of an array of float volume levels, from 0.0 to 1.0. These represent the incoming channels for the sound playing on the channel at the time.

*numlevels*

Number of floats to receive into the array. Maximum = the maximum number of input channels specified in [System::setSoftwareFormat](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This does not affect which speakers the sound is routed to. This can be used in conjunction with functions like [Channel::setPan](#), [Channel::setSpeakerMix](#), [Channel::setSpeakerLevels](#).

This function only scales the input channels from the sound.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setInputChannelMix](#)
- [Channel::setPan](#)
- [Channel::setSpeakerMix](#)

- [Channel::setSpeakerLevels](#)
- [System::setSoftwareFormat](#)

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# Channel::getLoopCount

Retrieves the current loop count for the specified channel.

## Syntax

```
FMOD_RESULT Channel::getLoopCount(  
    int * loopCount  
);
```

## Parameters

*loopCount*

Address of a variable that receives the number of times a channel will loop before stopping. 0 = oneshot. 1 = loop once then stop. -1 = loop forever. Default = -1

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function retrieves the **current** loop countdown value for the channel being played.

This means it will decrement until reaching 0, as it plays. To reset the value, use [Channel::setLoopCount](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setLoopCount](#)

# Channel::getLoopPoints

Retrieves the loop points for a channel.

## Syntax

```
FMOD_RESULT Channel::getLoopPoints (
    unsigned int * loopstart,
    FMOD_TIMEUNIT loopstarttype,
    unsigned int * loopend,
    FMOD_TIMEUNIT loopendtype
);
```

## Parameters

*loopstart*

Address of a variable to receive the loop start point. This point in time is played, so it is inclusive. Optional. Specify 0 or NULL to ignore.

*loopstarttype*

The time format used for the returned loop start point. See [FMOD\\_TIMEUNIT](#).

*loopend*

Address of a variable to receive the loop end point. This point in time is played, so it is inclusive. Optional. Specify 0 or NULL to ignore.

*loopendtype*

The time format used for the returned loop end point. See [FMOD\\_TIMEUNIT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_TIMEUNIT](#)
- [Channel::setLoopPoints](#)





# Channel::getMode

Retrieves the current mode bit flags for the current channel.

## Syntax

```
FMOD_RESULT Channel::getMode (
    FMOD_MODE * mode
);
```

## Parameters

*mode*

Address of an [FMOD\\_MODE](#) variable that receives the current mode for this channel.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setMode](#)

# Channel::getMute

Returns the current mute status of the channel.

## Syntax

```
FMOD_RESULT Channel::getMute (
    bool * mute
);
```

## Parameters

*mute*

true = channel is muted (silent), false = channel is at normal volume.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setMute](#)

# Channel::getPan

Returns the pan position of the channel.

## Syntax

```
FMOD_RESULT Channel::getPan(  
    float * pan  
);
```

## Parameters

*pan*

Address of a variable to receive the left/right pan level for the channel, from -1.0 to 1.0 inclusive. -1.0 = Full left, 1.0 = full right. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setPan](#)

# Channel::getPaused

Retrieves the paused state of the channel.

## Syntax

```
FMOD_RESULT Channel::getPaused(  
    bool * paused  
);
```

## Parameters

*paused*

Address of a variable that receives the current paused state. true = the sound is paused. false = the sound is not paused.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setPaused](#)

# Channel::getPosition

Returns the current PCM offset or playback position for the specified channel.

## Syntax

```
FMOD_RESULT Channel::getPosition(
    unsigned int * position,
    FMOD_TIMEUNIT postype
);
```

## Parameters

*position*

Address of a variable that receives the position of the sound.

*postype*

Time unit to retrieve into the position parameter. See [FMOD\\_TIMEUNIT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Certain timeunits do not work depending on the file format. For example [FMOD\\_TIMEUNIT\\_MODORDER](#) will not work with an mp3 file.

A PCM sample is a unit of measurement in audio that contains the data for one audible element of sound. 1 sample might be 16bit stereo, so 1 sample contains 4 bytes. 44,100 samples of a 44khz sound would represent 1 second of data.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setPosition](#)
- [FMOD\\_TIMEUNIT](#)
- [Sound::getLength](#)



# Channel::getPriority

Retrieves the current priority for this channel.

## Syntax

```
FMOD_RESULT Channel::getPriority (
    int * priority
);
```

## Parameters

*priority*

Address of a variable that receives the current channel priority. 0 to 256 inclusive. 0 = most important. 256 = least important. Default = 128.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setPriority](#)

# Channel::getReverbProperties

Retrieves the current reverb properties for this channel.

## Syntax

```
FMOD_RESULT Channel::getReverbProperties (
    FMOD_REVERB_CHANNELPROPERTIES * prop
);
```

## Parameters

*prop*

Address of a variable to receive the [FMOD\\_REVERB\\_CHANNELPROPERTIES](#) information.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setReverbProperties](#)
- [FMOD\\_REVERB\\_CHANNELPROPERTIES](#)



# Channel::getSpeakerLevels

Retrieves the current level settings from [Channel::setSpeakerLevels](#)?

## Syntax

```
FMOD_RESULT Channel::getSpeakerLevels (
    FMOD_SPEAKER speaker,
    float * levels,
    int numlevels
);
```

## Parameters

*speaker*

The speaker id to get the levels for. This can be cast to an integer if you are using a device with more than the pre-defined speaker range.

*levels*

Address of a variable that receives the current levels for the channel. This is an array of floating point values. The destination array size can be specified with the numlevels parameter.

*numlevels*

Number of floats in the destination array.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function does not return level values reflecting [Channel::setPan](#) or [Channel::setVolume](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setSpeakerLevels](#)
- [Channel::setPan](#)

- [\\_Channel::setVolume](#)

Version 4.12.03 Built on Feb 18, 2008

# Channel::getSpeakerMix

Sets the channel's speaker volume levels for each speaker individually.?

## Syntax

```
MOD ESU LTC h na l: ge s pake Mi x(  
    fba t * frontlft,  
    fba t * frontrght,  
    fba t * center,  
    fba t * lfe,  
    fba t * backlft,  
    fba t * backrght,  
    fba t * sidelft,  
    fba t * siderght  
);
```

## Parameters

*frontleft*

Address of a variable to receive the current volume level for this channel in the front left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

*frontright*

Address of a variable to receive the current volume level for this channel in the front right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

*center*

Address of a variable to receive the current volume level for this channel in the center speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

*lfe*

Address of a variable to receive the current volume level for this channel in the subwoofer speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

*backleft*

Address of a variable to receive the current volume level for this channel in the back left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

*backright*

Address of a variable to receive the current volume level for this channel in the back right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

*sideleft*

Address of a variable to receive the current volume level for this channel in the side left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

*sideright*

Address of a variable to receive the current volume level for this channel in the side right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume, up to 5.0 = 5x amplification.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

For 3D sound, the values set here are not representative of the 3d mix. For 3D sound this function is mainly for retrieving the LFE value if it was set by the user.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Channel:setSpeakerMix](#)

# Channel::getSpectrum

Retrieves the spectrum from the currently playing output signal for the current channel only.?

## Syntax

```
FMOD_RESULT Channel::getSpectrum(  
    float * spectrumarray,  
    int numvalues,  
    int channeloffset,  
    FMOD_DSP_FFT_WINDOW windowtype  
);
```

## Parameters

*spectrumarray*

Address of a variable that receives the spectrum data. This is an array of floating point values. Data will range is 0.0 to 1.0. Decibels =  $10.0f * (\text{float})\log_{10}(\text{val}) * 2.0f$ ; See remarks for what the data represents.

*numvalues*

Size of array in floating point values being passed to the function. Must be a power of 2. (ie 128/256/512 etc). Min = 64. Max = 8192.

*channeloffset*

Channel of the signal to analyze. If the signal is multichannel (such as a stereo output), then this value represents which channel to analyze. On a stereo signal 0 = left, 1 = right.

*windowtype*

"Pre-FFT" window method. This filters the PCM data before entering the spectrum analyzer to reduce transient frequency error for more accurate results. See [FMOD\\_DSP\\_FFT\\_WINDOW](#) for different types of fft window techniques possible and for a more detailed explanation.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The larger the numvalues, the more CPU the FFT will take. Choose the right value to trade off between accuracy / speed.

The larger the numvalues, the more 'lag' the spectrum will seem to inherit. This is because the FFT window size stretches the analysis back in time to what was already played. For example if the numvalues size happened to be 44100 and the output rate was 44100 it would be analyzing the past second of data, and giving you the average spectrum over that time period.

If you are not displaying the result in dB, then the data may seem smaller than it should be. To display it you may want to normalize the data - that is, find the maximum value in the resulting spectrum, and scale all values in the array by  $1 / \text{max}$ . (ie if the max was 0.5f, then it would become 1).

To get the spectrum for both channels of a stereo signal, call this function twice, once with `channeloffset = 0`, and again with `channeloffset = 1`. Then add the spectrums together and divide by 2 to get the average spectrum for both channels.

#### What the data represents.

To work out what each entry in the array represents, use this formula

$$\text{entry\_hz} = \text{output\_rate} / 2 / \text{num\_values}$$

The array represents amplitudes of each frequency band from 0hz to the nyquist rate. The nyquist rate is equal to the output rate divided by 2.

For example when FMOD is set to 44100hz output, the range of represented frequencies will be 0hz to 22049hz, a total of 22050hz represented.

If in the same example, 1024 was passed to this function as the numvalues, each entry's contribution would be as follows.

$$\text{entry\_hz} = (44100 / 2 / 1024)$$

$$\text{entry\_hz} = 21.53 \text{ hz}$$

**Note:** This function only displays data for sounds playing that were created with [FMOD\\_SOFTWARE](#). [FMOD\\_HARDWARE](#) based sounds are played using the sound card driver and are not accessible.

### Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

### See Also

- [FMOD\\_DSP\\_FFT\\_WINDOW](#)
- [System::getSpectrum](#)
- [ChannelGroup::getSpectrum](#)
- [System::getWaveData](#)

# Channel::getSystemObject

Retrieves the parent System object that was used to create this object.?

## Syntax

```
FMOD_RESULT Channel::getSystemObject(
    FMOD_SYSTEM **system
);
```

## Parameters

*system*

Address of a variable that receives the System object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::playSound](#)

# Channel::getUserData

Retrieves the user value that that was set by calling the [Channel::setUserData](#) function.?

## Syntax

```
FMOD_RESULT Channel::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [Channel::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

NOTE: If this channel was spawned by the event system then its user data field will be set, by the event system, to the event instance handle that spawned it. Use this function to go from an arbitrary channel back up to the event that owns it.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setUserData](#)



# Channel::getVolume

Retrieves the volume level for the channel.

## Syntax

```
FMOD_RESULT Channel::getVolume (
    float * volume
);
```

## Parameters

*volume*

Address of a variable to receive the channel volume level, from 0.0 to 1.0 inclusive. 0.0 = silent, 1.0 = full volume.  
Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setVolume](#)

# Channel::getWaveData

Retrieves a pointer to a block of PCM data that represents the currently playing waveform on this channel.  
?This function is useful for a very easy way to plot an oscilloscope.?

## Syntax

```
FMOD_RESULT Channel::getWaveData (
    float * wavearray ,
    int numvalues ,
    int channeloffset
);
```

## Parameters

*wavearray*

Address of a variable that receives the currently playing waveform data. This is an array of floating point values.

*numvalues*

Number of floats to write to the array. Maximum value = 16384.

*channeloffset*

Offset into multichannel data. Mono channels use 0. Stereo channels use 0 = left, 1 = right. More than stereo use the appropriate index.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This is the actual resampled pcm data window at the time the function is called.

Do not use this function to try and display the whole waveform of the sound, as this is more of a 'snapshot' of the current waveform at the time it is called, and could return the same data if it is called very quickly in succession. See the DSP API to capture a continual stream of wave data as it plays, or see [Sound::lock](#) / [Sound::unlock](#) if you want to simply display the waveform of a sound.

This function allows retrieval of left and right data for a stereo sound individually. To combine them into one signal, simply add the entries of each separate buffer together and then divide them by 2.

**Note:** This function only displays data for sounds playing that were created with [FMOD\\_SOFTWARE](#). [FMOD\\_HARDWARE](#) based sounds are played using the sound card driver and are not accessible.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getSpectrum](#)
- [ChannelGroup::getWaveData](#)
- [System::getWaveData](#)
- [Sound::lock](#)
- [Sound::unlock](#)

# Channel::isPlaying

Returns the playing state for the current channel.?

## Syntax

```
FMOD_RESULT Channel::isPlaying (
    bool * is_playing
);
```

## Parameters

*is\_playing*

Address of a variable that receives the current channel's playing status. true = the channel is currently playing a sound.  
false = the channel is not playing a sound.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::playSound](#)
- [System::playDSP](#)

# Channel::isVirtual

Returns the current channel's status of whether it is virtual (emulated) or not due to FMOD Ex's virtual channel management system.

## Syntax

```
FMOD_RESULT Channel::isVirtual(  
    bool * isVirtual  
);
```

## Parameters

*isVirtual*

Address of a variable that receives the current channel's virtual status. true = the channel is inaudible and currently being emulated at no cpu cost. false = the channel is a real hardware or software voice and should be audible.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Virtual channels are not audible, because there are no more real hardware or software channels available. If you are plotting virtual voices vs real voices graphically, and wondering why FMOD sometimes chooses seemingly random channels to be virtual that are usually far away, that is because they are probably silent. It doesn't matter which are virtual and which are not if they are silent. Virtual voices are not calculation on 'closest to listener' calculation, they are based on audibility. See the tutorial in the FMOD Ex documentation for more information on virtual channels.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::playSound](#)

# Channel::set3DAttributes

Sets the position and velocity of a 3d channel.

## Syntax

```
FMOD_RESULT Channel::set3DAttributes (
    const FMOD_VECTOR * pos,
    const FMOD_VECTOR * vel
);
```

## Parameters

*pos*

Position in 3D space of the channel. Specifying 0 / null will ignore this parameter.

*vel*

Velocity in 'distance units per second' in 3D space of the channel. See remarks. Specifying 0 / null will ignore this parameter.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A 'distance unit' is specified by [System::set3DSettings](#). By default this is set to meters which is a distance scale of 1.0.

For a stereo 3d sound, you can set the spread of the left/right parts in speaker space by using [Channel::set3DSpread](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::get3DAttributes](#)
- [FMOD\\_VECTOR](#)
- [System::set3DSettings](#)
- [Channel::set3DSpread](#)



# Channel::set3DConeOrientation

Sets the orientation of the sound projection cone.?

## Syntax

```
FMOD_RESULT Channel::set3DConeOrientation(
    FMOD_VECTOR *orientation
);
```

## Parameters

*orientation*

Pointer to an [FMOD\\_VECTOR](#) defining the coordinates of the sound cone orientation vector.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function has no effect unless the cone angle and cone outside volume have also been set to values other than the default.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::get3DConeOrientation](#)
- [Channel::set3DConeSettings](#)
- [Sound::set3DConeSettings](#)
- [FMOD\\_VECTOR](#)



# Channel::set3DConeSettings

Sets the inside and outside angles of the sound projection cone, as well as the volume of the sound outside the outside angle of the sound projection cone.

## Syntax

```
FMOD_RESULT Channel::set3DConeSettings (
    float insideconeangle,
    float outsideconeangle,
    float outsidevolume
);
```

## Parameters

*insideconeangle*

Inside cone angle, in degrees. This is the angle within which the sound is at its normal volume. Must not be greater than outsideconeangle. Default = 360.

*outsideconeangle*

Outside cone angle, in degrees. This is the angle outside of which the sound is at its outside volume. Must not be less than insideconeangle. Default = 360.

*outsidevolume*

Cone outside volume, from 0 to 1.0. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::get3DConeSettings](#)
- [Channel::set3DConeOrientation](#)
- [Sound::set3DConeSettings](#)



# Channel::set3DCustomRolloff

Point a channel to use a custom rolloff curve. Must be used in conjunction with [FMOD\\_3D\\_CUSTOMROLLOFF](#) flag to be activated.?

## Syntax

```
FMOD_RESULT Channel::set3DCustomRolloff(
    FMOD_VECTOR * points,
    int numpoints
);
```

## Parameters

*points*

An array of [FMOD\\_VECTOR](#) structures where x = distance and y = volume from 0.0 to 1.0. z should be set to 0.

*numpoints*

The number of points in the array.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

**Note!** This function does not duplicate the memory for the points internally. The pointer you pass to FMOD must remain valid until there is no more use for it.

Do not free the memory while in use, or use a local variable that goes out of scope while in use.

Points must be sorted by distance! Passing an unsorted list to FMOD will result in an error.

Set the points parameter to 0 or NULL to disable the points. If [FMOD\\_3D\\_CUSTOMROLLOFF](#) is set and the rolloff curve is 0, FMOD will revert to logarithmic curve rolloff.

Min and maxdistance are meaningless when [FMOD\\_3D\\_CUSTOMROLLOFF](#) is used and the values are ignored.

Here is an example of a custom array of points.

```
FMOD\_VECTOR Rcurve[ 3] =
{
    { 0.0f, 1.0f, 0.0f} ,
    { 2.0f, 0.2f, 0.0f} ,
    { 20.0f, 0.0f, 0.0f}
```

} ;

x represents the distance, y represents the volume. z is always 0.

Distances between points are linearly interpolated.

Note that after the highest distance specified, the volume in the last entry is used from that distance onwards.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MODE](#)
- [FMOD\\_VECTOR](#)
- [Channel::get3DCustomRolloff](#)
- [Sound::set3DCustomRolloff](#)
- [Sound::get3DCustomRolloff](#)

# Channel::set3DDopplerLevel

Sets the channel specific doppler scale for the channel.?

## Syntax

```
FMOD_RESULT channel::set3DDopplerLevel(
    float level
);
```

## Parameters

*level*

0 = No doppler. 1 = Normal doppler. 5 = max. Default = 1.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::get3DDopplerLevel](#)

# Channel::set3DMinMaxDistance

Sets the minimum and maximum audible distance for a channel.

?

## Syntax

```
FMOD_RESULT Channel::set3DMinMaxDistance (
    float mindistance,
    float maxdistance
);
```

## Parameters

*mindistance*

The channel's minimum volume distance in "units". See remarks for more on units.

*maxdistance*

The channel's maximum volume distance in "units". See remarks for more on units.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

MinDistance is the minimum distance that the sound emitter will cease to continue growing louder at (as it approaches the listener).

Within the mindistance it stays at the constant loudest volume possible. Outside of this mindistance it begins to attenuate.

MaxDistance is the distance a sound stops attenuating at. Beyond this point it will stay at the volume it would be at maxdistance units from the listener and will not attenuate any more.

MinDistance is useful to give the impression that the sound is loud or soft in 3d space. An example of this is a small quiet object, such as a bumblebee, which you could set a mindistance of to 0.1 for example, which would cause it to attenuate quickly and dissapear when only a few meters away from the listener.

Another example is a jumbo jet, which you could set to a mindistance of 100.0, which would keep the sound volume at max until the listener was 100 meters away, then it would be hundreds of meters more before it would fade out.

In summary, increase the mindistance of a sound to make it 'louder' in a 3d world, and decrease it to make it 'quieter' in a 3d world.

maxdistance is effectively obsolete unless you need the sound to stop fading out at a certain point. Do not adjust this from the default if you dont need to.

Some people have the confusion that maxdistance is the point the sound will fade out to, this is not the case.

A 'distance unit' is specified by [System::set3DSettings](#). By default this is set to meters which is a distance scale of 1.0.

The default units for minimum and maximum distances are 1.0 and 10000.0f.

Volume drops off at  $\text{mindistance} / \text{distance}$ .

To define the min and max distance per sound and not per channel use [Sound::set3DMinMaxDistance](#).

If [FMOD\\_3D\\_CUSTOMROLLOFF](#) is used, then these values are stored, but ignored in 3d processing.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Channel::get3DMinMaxDistance](#)
- [\\_System::set3DSettings](#)
- [\\_Sound::set3DMinMaxDistance](#)

# Channel::set3DOcclusion

Sets the EAX or software based occlusion factors for a channel. If the FMOD geometry engine is not being used, this function can be called to produce the same audible effects, just without the built in polygon processing. FMOD's internal geometry engine calls this function.?

## Syntax

```
FMOD_RESULT Channel::set3DOcclusion(  
    float directOcclusion,  
    float reverbOcclusion  
);
```

## Parameters

*directOcclusion*

Occlusion factor for a voice for the direct path. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0.

*reverbOcclusion*

Occlusion factor for a voice for the reverb mix. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

With EAX based sound cards and [FMOD\\_HARDWARE](#) based sounds, this will attenuate the sound using frequency filtering.

With non EAX sounds, then the volume is simply attenuated by the directOcclusion factor.

If FMOD\_INIT\_OCCLUSION\_LOWPASS is specified, [FMOD\\_SOFTWARE](#) based sounds will also use frequency filtering, with a small CPU hit.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::get3DOcclusion](#)
- [ChannelGroup::set3DOcclusion](#)





# Channel::set3DPanLevel

Sets how much the 3d engine has an effect on the channel, versus that set by [Channel::setPan](#), [Channel::setSpeakerMix](#), [Channel::setSpeakerLevels](#).

?

## Syntax

```
FMOD_RESULT channel::set3DPanLevel(
    float level
);
```

## Parameters

*level*

1 = Sound pans and attenuates according to 3d position. 0 = Attenuation is ignored and pan/speaker levels are defined by [Channel::setPan](#), [Channel::setSpeakerMix](#), [Channel::setSpeakerLevels](#). Default = 1 (all by 3D position).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Only affects sounds created with [FMOD\\_SOFTWARE](#) and [FMOD\\_3D](#).

Useful for morphing a sound between 3D and 2D. This is most common in volumetric sound, when the sound goes from directional, to 'all around you' (and doesn't pan according to listener position/direction).

FMOD\_INIT\_SOFTWARE\_HRTF is also interpolated to be 'off' if level = 0, so that you do not get a muffling effect based on location when the sound is supposed to be effectively 2D.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::get3DPanLevel](#)
- [Channel::setSpeakerMix](#)
- [Channel::setPan](#)
- [Channel::setSpeakerLevels](#)



# Channel::set3DSpread

Sets the spread of a 3d stereo or multichannel sound in speaker space.

?Normally a 3d sound is aimed at one position in a speaker array depending on the 3d position, to give it direction.

Left and right parts of a stereo sound for example are consequently summed together and become 'mono'.

?When increasing the 'spread' of a sound, the left and right parts of a stereo sound rotate away from their original position, to give it more 'stereoness'. The rotation of the sound channels are done in 'speaker space'.?

## Syntax

```
FMOD_RESULT Channel::set3DSpread(  
    float angle  
) ;
```

## Parameters

*angle*

Spread angle for stereo sounds and above. 0 = all sound channels are located at the same speaker location and is 'mono'. 360 = all subchannels are located at the opposite speaker location to the speaker location that it should be according to 3D position. Default = 0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Only affects sounds created with [FMOD\\_SOFTWARE](#).

By default, if a stereo sound was played in 3d, and it was directly in front of you, the left and right part of the stereo sound would be summed into the center speaker (on a 5.1 setup), making it sound mono.

This function lets you control the speaker spread of a stereo (and above) sound within the speaker array, to separate the left right part of a stereo sound for example.

In the above case, in a 5.1 setup, specifying a spread of 90 degrees would put the left part of the sound in the front left speaker, and the right part of the sound in the front right speaker. This stereo separation remains in tact as the listener rotates and the sound moves around the speakers.

To summarize (for a stereo sound).

1. A spread angle of 0 makes the stereo sound mono at the point of the 3d emitter.
2. A spread angle of 90 makes the left part of the stereo sound place itself at 45 degrees to the left and the right part 45 degrees to the right.
3. A spread angle of 180 makes the left part of the stereo sound place itself at 90 degrees to the left and the right part 90 degrees to the right.
4. A spread angle of 360 makes the stereo sound mono at the opposite speaker location to where the 3d emitter should be located (by moving the left part 180 degrees left and the right part 180 degrees right). So in this case, behind you when the sound should be in front of you!

Multichannel sounds with channel counts greater than stereo have their sub-channels spread evenly through the

specified angle. For example a 6 channel sound over a 90 degree spread has each subchannel located 15 degrees apart from each other in the speaker array.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_Channel::get3DSpread](#)
- [\\_FMOD\\_MODE](#)

# Channel::setCallback

Sets a callback for a channel for a specific event.?

## Syntax

```
FMOD_RESULT Channel::setCallback (
    FMOD_CHANNEL_CALLBACK_TYPE type,
    FMOD_CHANNEL_CALLBACK callback,
    int command
);
```

## Parameters

*type*

The callback type, for example an 'end of sound' callback.

*callback*

Pointer to a callback to receive the event when it happens.

*command*

The callback parameter. This has a different meaning for each type of callback type.

**FMOD\_CHANNEL\_CALLBACK\_END** - This parameter has no effect.

**FMOD\_CHANNEL\_CALLBACK\_VIRTUALVOICE** - This parameter has no effect.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Currently callbacks are driven by [System::update](#) and will only occur when this function is called. This has the main advantage of far less complication due to thread issues, and allows all FMOD commands, including loading sounds and playing new sounds from the callback.

It also allows any type of sound to have an end callback, no matter what it is. The only disadvantage is that callbacks are not asynchronous and are bound by the latency caused by the rate the user calls the update command.

Callbacks are stdcall. Use F\_CALLBACK inbetween your return type and function name.

Example:

```
FMOD_RESULT CALLBACK mycallback ( FMOD_CHANNEL_CALLBACK_TYPE type,
    int command, unsigned int commanddata1, unsigned int commanddata2
){
```

```
    FMOD_Channel_t *channel = ( FMOD_Channel_t *)channel;
```

```
// More code goes here.
```

```
return FMOD\_OK;
```

```
}
```

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System:update](#)
- [FMOD\\_CHANNEL\\_CALLBACK](#)
- [FMOD\\_CHANNEL\\_CALLBACKTYPE](#)

# Channel::setChannelGroup

Sets a channel to belong to a specified channel group. A channelgroup can contain many channels.  
?

## Syntax

```
FMOD_RESULT Channel::setChannelGroup(  
    FMOD_CHANNELGROUP *channelgroup  
);
```

## Parameters

*channelgroup*

Pointer to a ChannelGroup object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Setting a channel to a channel group removes it from any previous group, it does not allow sharing of channel groups.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getChannelGroup](#)



# Channel::setDelay

Sets a delay before the sound is audible and after the sound ends.?

## Syntax

```
FMOD_RESULT Channel::setDelay (
    unsigned int startDelay,
    unsigned int endDelay
);
```

## Parameters

*startDelay*

The delay in milliseconds before the sound starts. Currently not implemented yet.

*endDelay*

The delay in milliseconds after the sound stops before the channel actually stops processing. [Channel::isPlaying](#) will remain true until this delay has passed even though the sound itself has stopped playing.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Setting a delay after a sound ends is sometimes useful to prolong the sound, even though it has stopped, so that DSP effects can trail out, or render the last of their tails. (for example an echo or reverb effect).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getDelay](#)
- [Channel::isPlaying](#)



# Channel::setFrequency

Sets the channel's frequency or playback rate, in HZ.?

## Syntax

```
FMOD_RESULT Channel::setFrequency (
    float frequency
);
```

## Parameters

*frequency*

A frequency value in HZ. This value can also be negative to play the sound backwards (negative frequencies allowed with [FMOD\\_SOFTWARE](#) based non-stream sounds only). DirectSound hardware voices have limited frequency range on some soundcards. Please see remarks for more on this.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When a sound is played, it plays at the default frequency of the sound which can be set by [Sound::setDefault](#). For most file formats, the volume is determined by the audio format.

Frequency limitations for sounds created with [FMOD\\_HARDWARE](#) in DirectSound.

Every hardware device has a minimum and maximum frequency. This means setting the frequency above the maximum and below the minimum will have no effect.

FMOD clamps frequencies to these values when playing back on hardware, so if you are setting the frequency outside of this range, the frequency will stay at either the minimum or maximum.

Note that [FMOD\\_SOFTWARE](#) based sounds do not have this limitation.

To find out the minimum and maximum value before initializing FMOD (maybe to decide whether to use a different soundcard, output mode, or drop back fully to software mixing), you can use the [System::getDriverCaps](#) function.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getFrequency](#)

- [\\_System::getDriverCaps](#)
- [\\_Sound::setDefaults](#)

Version 4.12.03 Built on Feb 18, 2008

# Channel::setInputChannelMix

Sets the incoming levels in a sound. This means if you have a multichannel sound you can turn channels on and off. A mono sound has 1 input channel, a stereo has 2, etc. It depends on what type of sound is playing on the channel at the time.

## Syntax

```
FMOD_RESULT Channel::setInputChannelMix(
    float * levels,
    int numlevels
);
```

## Parameters

*levels*

Array of float volume levels, from 0.0 to 1.0. These represent the incoming channels for the sound playing on the channel at the time.

*numlevels*

Number of floats in the array. Maximum = the maximum number of input channels specified in [System::setSoftwareFormat](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getInputChannelMix](#)
- [Channel::setPan](#)
- [Channel::setSpeakerMix](#)
- [Channel::setSpeakerLevels](#)
- [System::setSoftwareFormat](#)

# Channel::setLoopCount

Sets a channel to loop a specified number of times before stopping.?

## Syntax

```
FMOD_RESULT Channel::setLoopCount(  
    int loopcount  
);
```

## Parameters

*loopcount*

Number of times to loop before stopping. 0 = oneshot. 1 = loop once then stop. -1 = loop forever. Default = -1

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function does not affect [FMOD\\_HARDWARE](#) based sounds that are not streamable.

FMOD\_SOFTWARE based sounds or any type of sound created with System::CreateStream or [FMOD\\_CREATESTREAM](#) will support this function.

Issues with streamed audio. (Sounds created with with [System::createStream](#) or [FMOD\\_CREATESTREAM](#)). When changing the loop count, sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#) may already have been pre-buffered and executed their loop logic ahead of time, before this call was even made.

This is dependant on the size of the sound versus the size of the stream *decode* buffer. See [FMOD\\_CREATESOUNDEXINFO](#).

If this happens, you may need to reflush the stream buffer. To do this, you can call [Channel::setPosition](#) which forces a reflush of the stream buffer.

Note this will usually only happen if you have sounds or looppoints that are smaller than the stream decode buffer size. Otherwise you will not normally encounter any problems.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getLoopCount](#)

- [\\_Channel::setPosition](#)
- [\\_System::createStream](#)
- [FMOD\\_CREATEINDEXINFO](#)
- [FMOD\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008

# Channel::setLoopPoints

Sets the loop points within a channel.?

## Syntax

```
FMOD_RESULT Channel::setLoopPoints (
    unsigned int loopstart,
    FMOD_TIMEUNIT loopstarttype,
    unsigned int loopend,
    FMOD_TIMEUNIT loopendtype
);
```

## Parameters

*loopstart*

The loop start point. This point in time is played, so it is inclusive.

*loopstarttype*

The time format used for the loop start point. See [FMOD\\_TIMEUNIT](#).

*loopend*

The loop end point. This point in time is played, so it is inclusive.

*loopendtype*

The time format used for the loop end point. See [FMOD\\_TIMEUNIT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Not supported by static sounds created with [FMOD\\_HARDWARE](#).

Supported by sounds created with [FMOD\\_SOFTWARE](#), or sounds of any type (hardware or software) created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#).

If a sound was 1000ms long and you wanted to loop the whole sound, loopstart would be 0, and loopend would be 999,

not 1000.

If loop end is smaller or equal to loop start, it will result in an error.

If loop start or loop end is larger than the length of the sound, it will result in an error.

Issues with streamed audio. (Sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#)). When changing the loop points, sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#) may already



have been pre-buffered and executed their loop logic ahead of time, before this call was even made.

This is dependant on the size of the sound versus the size of the stream *decode* buffer. See

[FMOD\\_CREATEINDEXINFO](#).

If this happens, you may need to reflush the stream buffer. To do this, you can call `Channel::setPosition` which forces a reflush of the stream buffer.

Note this will usually only happen if you have sounds or looppoints that are smaller than the stream decode buffer size. Otherwise you will not normally encounter any problems.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_TIMEUNIT](#)
- [FMOD\\_MODE](#)
- [Channel::getLoopPoints](#)
- [Channel::setLoopCount](#)
- [System::createStream](#)
- [System::setStreamBufferSize](#)
- [FMOD\\_CREATEINDEXINFO](#)

# Channel::setMode

Changes some attributes for a channel based on the mode passed in.?

## Syntax

```
FMOD_RESULT Channel::setMode(  
    FMOD_MODE mode  
);
```

## Parameters

*mode*

Mode bits to set.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Flags supported:

FMOD\_LOOP\_OFF

FMOD\_LOOP\_NORMAL

FMOD\_LOOP\_BIDI (only works with sounds created with [FMOD\\_SOFTWARE](#). Otherwise it will behave as [FMOD\\_LOOP\\_NORMAL](#))

FMOD\_3D\_HEADRELATIVE

FMOD\_3D\_WORLDRELATIVE

FMOD\_2D (see notes for win32 hardware voices)

FMOD\_3D (see notes for win32 hardware voices)

FMOD\_3D\_LOGROLLOFF

FMOD\_3D\_LINEARROLLOFF

FMOD\_3D\_CUSTOMROLLOFF

FMOD\_3D\_IGNOREGEOMETRY

FMOD\_DONTRESTOREVIRTUAL

Issues with streamed audio. (Sounds created with with [System::createStream](#) or [FMOD\\_CREATESTREAM](#)). When changing the loop mode, sounds created with [System::createStream](#) or [FMOD\\_CREATESTREAM](#) may already have been pre-buffered and executed their loop logic ahead of time, before this call was even made.

This is dependant on the size of the sound versus the size of the stream *decode* buffer. See

[FMOD\\_CREATESOUNDEXINFO](#).

If this happens, you may need to reflush the stream buffer. To do this, you can call [Channel::setPosition](#) which forces a reflush of the stream buffer.

Note this will usually only happen if you have sounds or looppoints that are smaller than the stream decode buffer size. Otherwise you will not normally encounter any problems.

**Win32** [FMOD\\_HARDWARE](#) note. Under DirectSound, you cannot change the loop mode of a channel while it is

playing. You must use [Sound::setMode](#) or pause the channel to get this to work.

**Win32** [FMOD\\_HARDWARE](#) note. Under DirectSound, you cannot change the mode of a channel between [FMOD\\_2D](#) and [FMOD\\_3D](#). If this is a problem create the sound as [FMOD\\_3D](#) initially, and use [FMOD\\_3D\\_HEADRELATIVE](#) and [FMOD\\_3D\\_WORLDRELATIVE](#). Alternatively just use [FMOD\\_SOFTWARE](#).

If [FMOD\\_3D\\_IGNOREGEOMETRY](#) is not specified, the flag will be cleared if it was specified previously.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MODE](#)
- [Channel::getMode](#)
- [Channel::setPosition](#)
- [Sound::setMode](#)
- [System::createStream](#)
- [System::setStreamBufferSize](#)
- [FMOD\\_CREATEINDEXINFO](#)

# Channel::setMute

Mutes / un-mutes a channel, effectively silencing it or returning it to its normal volume.?

## Syntax

```
FMOD_RESULT Channel::setMute(  
    bool mute  
);
```

## Parameters

*mute*

true = channel becomes muted (silent), false = channel returns to normal volume.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a channel belongs to a muted channelgroup, it will stay muted regardless of the channel mute state. The channel mute state will still be reflected internally though, ie [Channel::getMute](#) will still return the value you set. If the channelgroup has mute set to false, this function will become effective again.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getMute](#)
- [ChannelGroup::setMute](#)

# Channel::setPan

Sets a channels pan position linearly.?

## Syntax

```
FMOD_RESULT Channel::setPan(  
    float pan  
);
```

## Parameters

*pan*

A left/right pan level, from -1.0 to 1.0 inclusive. -1.0 = Full left, 0.0 = center, 1.0 = full right. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function only works on sounds created with [FMOD\\_2D](#). 3D sounds are not pannable and will return [FMOD\\_ERR\\_NEEDS2D](#).

Only sounds that are mono or stereo can be panned. Multichannel sounds (ie >2 channels) cannot be panned. Mono sounds are panned from left to right using constant power panning (non linear fade). This means when pan = 0.0, the balance for the sound in each speaker is 71% left and 71% right, not 50% left and 50% right. This gives (audibly) smoother pans.

Stereo sounds have each left/right value faded up and down according to the specified pan position. This means when pan = 0.0, the balance for the sound in each speaker is 100% left and 100% right. When pan = -1.0, only the left channel of the stereo sound is audible, when pan = 1.0, only the right channel of the stereo sound is audible.

Panning does not work if the speaker mode is [FMOD\\_SPEAKERMODE\\_RAW](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getPan](#)
- [FMOD\\_SPEAKERMODE](#)

Version 4.12.03 Built on Feb 18, 2008

# Channel::setPaused

Sets the paused state of the channel.

## Syntax

```
FMOD_RESULT h na l: se tPause d(  
    bo l pause d  
);
```

## Parameters

*paused*

Paused state to set. true = channel is paused. false = channel is unpaused.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a channel belongs to a paused channelgroup, it will stay paused regardless of the channel pause state. The channel pause state will still be reflected internally though, ie [Channel::getPaused](#) will still return the value you set. If the channelgroup has paused set to false, this function will become effective again.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getPaused](#)
- [ChannelGroup::setPaused](#)

# Channel::setPosition

Sets the current playback position for the currently playing sound to the specified PCM offset.?

## Syntax

```
FMOD_RESULT Channel::setPosition(  
    unsigned int position,  
    FMOD_TIMEUNIT position_type  
);
```

## Parameters

*position*

Position of the channel to set in units specified in the postype parameter.

*postype*

Time unit to set the channel position by. See [FMOD\\_TIMEUNIT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Certain timeunits do not work depending on the file format. For example [FMOD\\_TIMEUNIT\\_MODORDER](#) will not work with an mp3 file.

Note that if you are calling this function on a stream, it has to possibly reflush its buffer to get zero latency playback when it resumes playing, therefore it could potentially cause a stall or take a small amount of time to do this.

**Warning!** Using a VBR source that does not have an associated seek table or seek information (such as MP3 or MOD/S3M/XM/IT) may cause inaccurate seeking if you specify [FMOD\\_TIMEUNIT\\_MS](#) or [FMOD\\_TIMEUNIT\\_PCM](#).

If you want FMOD to create a pcm vs bytes seek table so that seeking is accurate, you will have to specify [FMOD\\_ACCURATETIME](#) when loading or opening the sound. This means there is a slight delay as FMOD scans the whole file when loading the sound to create this table.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris



## See Also

- [\\_Channel::getPosition](#)
- [FMOD\\_TIMEUNIT](#)
- [FMOD\\_MODE](#)
- [\\_Sound::getLength](#)

Version 4.12.03 Built on Feb 18, 2008

# Channel::setPriority

Sets the priority for a channel after it has been played. A sound with a higher priority than another sound will not be stolen or made virtual by that sound.?

## Syntax

```
FMOD_RESULT Channel::setPriority (
    int priority
);
```

## Parameters

*priority*

priority for the channel. 0 to 256 inclusive. 0 = most important. 256 = least important. Default = 128.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Priority will make a channel more important or less important than its counterparts. When virtual channels are in place, by default the importance of the sound (whether it is audible or not when more channels are playing than exist) is based on the volume, or audibility of the sound. This is determined by distance from the listener in 3d, the volume set with [Channel::setVolume](#), channel group volume, and geometry occlusion factors. To make a quiet sound more important, so that it isn't made virtual by louder sounds, you can use this function to increase its importance, and keep it audible.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getPriority](#)
- [Channel::setVolume](#)

# Channel::setReverbProperties

Sets the channel specific reverb properties, including things like wet/dry mix.?

## Syntax

```
FMOD_RESULT Channel::setReverbProperties (
    const FMOD_REVERB_CHANNELPROPERTIES * prop
);
```

## Parameters

*prop*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

With [FMOD\\_HARDWARE](#) on Windows using EAX, the reverb will only work on [FMOD\\_3D](#) based sounds. [FMOD\\_SOFTWARE](#) does not have this problem and works on [FMOD\\_2D](#) and [FMOD\\_3D](#) based sounds.

On PlayStation 2, the 'Room' parameter is the only parameter supported. The hardware only allows 'on' or 'off', so the reverb will be off when 'Room' is -10000 and on for every other value.

On Xbox, it is possible to apply reverb to [FMOD\\_2D](#) and [FMOD\\_HARDWARE](#) based voices using this function. By default reverb is turned off for [FMOD\\_2D](#) hardware based voices.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getReverbProperties](#)
- [FMOD\\_REVERB\\_CHANNELPROPERTIES](#)

# Channel::setSpeakerLevels

Sets the incoming sound levels for a particular speaker.?

## Syntax

```
FMOD_RESULT Channel::setSpeakerLevels (
    FMOD_SPEAKER speaker,
    float * levels,
    int numlevels
);
```

## Parameters

*speaker*

The target speaker to modify the levels for. This can be cast to an integer if you are using [FMOD\\_SPEAKERMODE\\_RAW](#) and want to access up to 15 speakers (output channels).

*levels*

An array of floating point numbers from 0.0 to 1.0 representing the volume of each input channel of a sound. See remarks for more.

*numlevels*

The number of floats within the levels parameter being passed to this function. In the case of the above mono or stereo sound, 1 or 2 could be used respectively. If the sound being played was an 8 channel multichannel sound then 8 levels would be used.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

As an example of usage of this function, if the sound played on this speaker was mono, only 1 level would be needed.

If the sound played on this channel was stereo, then an array of 2 floats could be specified. For example { 0, 1 } on a channel playing a stereo sound would mute the left part of the stereo sound when it is played on this speaker.

**Note!** In [FMOD\\_SPEAKERMODE\\_MONO](#) it is preferable to use the alias [FMOD\\_SPEAKER\\_MONO](#).

Only speakers that are usable with the current speaker mode will be accepted. Anything else will return [FMOD\\_ERR\\_INVALID\\_SPEAKER](#).

Under [FMOD\\_SPEAKERMODE\\_RAW](#), the 'speaker' parameter can be cast to an integer and used as a raw speaker index, disregarding FMOD's speaker mappings.

**Warning.** This function will allocate memory for the speaker level matrix and attach it to the channel. If you prefer not to have a dynamic memory allocation done at this point use [Channel::setSpeakerMix](#) instead.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getSpeakerLevels](#)
- [Channel::setSpeakerMix](#)
- [FMOD\\_SPEAKERMODE](#)
- [FMOD\\_SPEAKER](#)

# Channel::setSpeakerMix

Sets the channel's speaker volume levels for each speaker individually.?

## Syntax

```
MOD_RESULT channel::setSpeakerMix(  
    float frontLeft,  
    float frontRight,  
    float center,  
    float lfe,  
    float backLeft,  
    float backRight,  
    float sideLeft,  
    float sideRight  
);
```

## Parameters

*frontLeft*

Volume level for this channel in the front left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, 5.0 = 5x amplification.

*frontRight*

Volume level for this channel in the front right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, up to 5.0 = 5x amplification.

*center*

Volume level for this channel in the center speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, 5.0 = 5x amplification.

*lfe*

Volume level for this channel in the subwoofer speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, 5.0 = 5x amplification.

*backLeft*

Volume level for this channel in the back left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, 5.0 = 5x amplification.

*backRight*

Volume level for this channel in the back right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, 5.0 = 5x amplification.

*sideLeft*

Volume level for this channel in the side left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, 5.0 = 5x amplification.

*sideright*

Volume level for this channel in the side right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = normal volume, 5.0 = 5x amplification.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function only fully works on sounds created with [FMOD\\_2D](#) and [FMOD\\_SOFTWARE](#). [FMOD\\_3D](#) based sounds only allow setting of LFE channel, as all other speaker levels are calculated by FMOD's 3D engine.

Speakers specified that don't exist will simply be ignored.

For more advanced speaker control, including sending the different channels of a stereo sound to arbitrary speakers, see [Channel::setSpeakerLevels](#).

This function allows amplification! You can go up to 5 times the volume of a normal sound, but warning this may cause clipping/distortion! Useful for LFE boosting.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getSpeakerMix](#)
- [Channel::setSpeakerLevels](#)
- [FMOD\\_SPEAKERMODE](#)

# Channel::setUserData

Sets a user value that the Channel object will store internally. Can be retrieved with [Channel::getUserData](#)?

## Syntax

```
FMOD_RESULT Channel::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the Channel object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [Channel::getUserData](#) would help in the identification of the object.

NOTE: If this channel was spawned by the event system then its user data field will be set, by the event system, to the event instance handle that spawned it and this function should NOT be called.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getUserData](#)



# Channel::setVolume

Sets the volume for the channel linearly.?

## Syntax

```
FMOD_RESULT Channel::setVolume (
    float volume
);
```

## Parameters

*volume*

A linear volume level, from 0.0 to 1.0 inclusive. 0.0 = silent, 1.0 = full volume. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When a sound is played, it plays at the default volume of the sound which can be set by [Sound::setDefaults](#).

For most file formats, the volume is determined by the audio format.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::getVolume](#)
- [ChannelGroup::setVolume](#)
- [Sound::setDefaults](#)

# Channel::stop

Stops the channel from playing. Makes it available for re-use by the priority system.?

## Syntax

```
FMOD_RESULT Channel::stop();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::playSound](#)

# ChannelGroup Interface

[ChannelGroup::addDSP](#)  
[ChannelGroup::addGroup](#)  
[ChannelGroup::get3DOcclusion](#)  
[ChannelGroup::getChannel](#)  
[ChannelGroup::getDSPHead](#)  
[ChannelGroup::getGroup](#)  
[ChannelGroup::getMute](#)  
[ChannelGroup::getName](#)  
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[ChannelGroup::getSpectrum](#)  
[ChannelGroup::getSystemObject](#)  
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[ChannelGroup::getVolume](#)  
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[ChannelGroup::override3DAttributes](#)  
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[ChannelGroup::overrideVolume](#)  
[ChannelGroup::release](#)  
[ChannelGroup::set3DOcclusion](#)  
[ChannelGroup::setMute](#)  
[ChannelGroup::setPaused](#)  
[ChannelGroup::setPitch](#)  
[ChannelGroup::setUserData](#)  
[ChannelGroup::setVolume](#)  
[ChannelGroup::stop](#)

# ChannelGroup::addDSP

Adds a DSP effect to this channelgroup, affecting all channels that belong to it. Because it is a submix, only one instance of the effect is added, and all subsequent channels are affected.?

## Syntax

```
FMOD_RESULT ChannelGroup::addDSP(
    FMOD_DSP * dsp
);
```

## Parameters

*dsp*

Pointer to the dsp effect to add. This can be created with [System::createDSP](#), [System::createDSPByType](#), [System::createDSPByIndex](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is a wrapper function to insert a DSP unit at the top of the channel group DSP chain.

It disconnects the head unit from its input, then inserts the unit at the head and reconnects the previously disconnected input back as an input to the new unit.

It is effectively the following code.

```
int main()
{
    ChannelGroup *cg;
    DSP *dsp;
    if (main() > 1)
    {
        return FMOD_ERR_DSP_INVALID;
    }
    dsp = cg->getDSP(0);
    dsp->disconnectFrom(dsp);
    dsp->addInput(dsp);
    dsp->addInput(dsp);
    dsp->setVolume(1.0f);
}
```

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable,

PlayStation 3, Wii, Solaris

#### See Also

- [\\_ChannelGroup::getDSPHead](#)
- [\\_System::createDSP](#)
- [\\_System::createDSPByType](#)
- [\\_System::createDSPByIndex](#)
- [\\_System::getMasterChannelGroup](#)
- [\\_System::createChannelGroup](#)
- [\\_System::addDSP](#)
- [\\_Channel::addDSP](#)
- [\\_DSP::remove](#)

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# ChannelGroup::addGroup

Adds a channel group as a child of the current channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::addGroup(
    FMOD_CHANNELGROUP *group
);
```

## Parameters

*group*

channel group to add as a child.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getNumGroups](#)
- [ChannelGroup::getGroup](#)

# ChannelGroup::get3DOcclusion

Retrieves the master occlusion factors for the channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::get3DOcclusion(  
    float * directOcclusion,  
    float * reverbOcclusion  
);
```

## Parameters

*directOcclusion*

Address of a variable that receives the occlusion factor for the direct path. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0. Optional. Specify 0 or NULL to ignore.

*reverbOcclusion*

Address of a variable that receives the occlusion factor for the reverb mix. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::set3DOcclusion](#)
- [Channel::set3DOcclusion](#)
- [Channel::get3DOcclusion](#)
- [System::getMasterChannelGroup](#)

# ChannelGroup::getChannel

Retrieves the a handle to a channel from the current channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::getChannel(  
    int index,  
    FMOD_CHANNEL** channel  
);
```

## Parameters

*index*

Index of the channel inside the channel group, from 0 to the number of channels returned by [ChannelGroup::getNumChannels](#).

*channel*

Address of a variable to receive a pointer to a Channel object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getNumChannels](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)



# ChannelGroup::getDSPHead

Retrieves the DSP unit responsible for this channel group. When channels are submixed to this channel group, this is the DSP unit they target.?

## Syntax

```
FMOD_RESULT ChannelGroup::getDSPHead(
    FMOD_DSP** dsp
);
```

## Parameters

*dsp*

Address of a variable to receive the pointer to the head DSP unit for this channel group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use this unit if you wish to connect custom DSP units to the channelgroup or filter the channels in the channel group by inserting filter units between this one and the incoming channel mixer unit.

Read the tutorial on DSP if you wish to know more about this. It is not recommended using this if you do not understand how the FMOD Ex DSP network is connected.

Alternatively you can simply add effects by using [ChannelGroup::addDSP](#) which does the connection / disconnection work for you.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::addDSP](#)
- [System::createDSP](#)
- [System::createDSPByType](#)
- [System::createDSPByIndex](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)



# ChannelGroup::getGroup

Retrieves a handle to a specified sub channelgroup.?

## Syntax

```
FMOD_RESULT ChannelGroup::getGroup(
    int index,
    FMOD_CHANNELGROUP** group
);
```

## Parameters

*index*

Index to specify which sub channelgroup to receive.

*group*

Address of a variable to receive a pointer to a channelgroup.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getNumGroups](#)
- [ChannelGroup::getParentGroup](#)
- [ChannelGroup::addGroup](#)

# ChannelGroup::getMute

Retrieves the mute state of a ChannelGroup.?

## Syntax

```
FMOD_RESULT ChannelGroup::getMute (
    bool * mute
);
```

## Parameters

*mute*

Address of a variable to receive the pause state of the channelgroup.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::setMute](#)
- [Channel::setMute](#)
- [Channel::getMute](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)

# ChannelGroup::getName

Retrieves the name of the channelgroup. The name is set when the group is created.?

## Syntax

```
FMOD_RESULT ChannelGroup::getName (
    char * name,
    int nameLen
);
```

## Parameters

*name*

Address of a variable that receives the name of the channel group.

*namelen*

Length in bytes of the target buffer to receive the string.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)

# ChannelGroup::getNumChannels

Retrieves the current number of assigned channels to this channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::getNumChannels (
    int * numchannels
);
```

## Parameters

*numchannels*

Address of a variable to receive the current number of assigned channels in this channel group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use this function to enumerate the channels within the channel group. You can then use [ChannelGroup::getChannel](#) to retrieve each individual channel.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getChannel](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)

# ChannelGroup::getNumGroups

Retrieves the number of sub groups under this channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::getNumGroups (
    int * numgroups
);
```

## Parameters

*numgroups*

Address of a variable to receive the number of channel groups within this channel group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getGroup](#)
- [ChannelGroup::addGroup](#)

# ChannelGroup::getParentGroup

Retrieves a handle to this channelgroup's parent channelgroup.?

## Syntax

```
FMOD_RESULT ChannelGroup::getParentGroup(
    FMOD_CHANNELGROUP **group
);
```

## Parameters

*group*

Address of a variable to receive a pointer to a channelgroup.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getNumGroups](#)
- [ChannelGroup::getGroup](#)



# ChannelGroup::getPaused

Retrieves the pause state of a ChannelGroup.?

## Syntax

```
FMOD_RESULT ChannelGroup::getPaused(  
    bool * paused  
);
```

## Parameters

*paused*

Address of a variable to receive the pause state of the channelgroup.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::setPaused](#)
- [Channel::setPaused](#)
- [Channel::getPaused](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)

# ChannelGroup::getPitch

Retrieves the master pitch level for the channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::getPitch(  
    float * pitch  
);
```

## Parameters

*pitch*

Address of a variable to receive the channel group pitch value, from 0.0 to 10.0 inclusive. 0.0 = silent, 1.0 = full volume. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::setPitch](#)
- [ChannelGroup::overrideFrequency](#)
- [System::getMasterChannelGroup](#)

# ChannelGroup::getSpectrum

Retrieves the spectrum from the currently playing channels assigned to this channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::getSpectrum(  
    float * spectrumarray,  
    int numvalues,  
    int channeloffset,  
    FMOD_DSP_FFT_WINDOW windowtype  
);
```

## Parameters

*spectrumarray*

Address of a variable that receives the spectrum data. This is an array of floating point values. Data will range is 0.0 to 1.0. Decibels =  $10.0f * (\text{float})\log_{10}(\text{val}) * 2.0f$ ; See remarks for what the data represents.

*numvalues*

Size of array in floating point values being passed to the function. Must be a power of 2. (ie 128/256/512 etc). Min = 64. Max = 8192.

*channeloffset*

Channel of the signal to analyze. If the signal is multichannel (such as a stereo output), then this value represents which channel to analyze. On a stereo signal 0 = left, 1 = right.

*windowtype*

"Pre-FFT" window method. This filters the PCM data before entering the spectrum analyzer to reduce transient frequency error for more accurate results. See [FMOD\\_DSP\\_FFT\\_WINDOW](#) for different types of fft window techniques possible and for a more detailed explanation.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The larger the numvalues, the more CPU the FFT will take. Choose the right value to trade off between accuracy / speed.

The larger the numvalues, the more 'lag' the spectrum will seem to inherit. This is because the FFT window size stretches the analysis back in time to what was already played. For example if the window size happened to be 44100 and the output rate was 44100 it would be analyzing the past second of data, and giving you the average spectrum over that time period.

If you are not displaying the result in dB, then the data may seem smaller than it should be. To display it you may want to normalize the data - that is, find the maximum value in the resulting spectrum, and scale all values in the array by  $1 / \text{max}$ . (ie if the max was 0.5f, then it would become 1).

To get the spectrum for both channels of a stereo signal, call this function twice, once with `channeloffset = 0`, and again with `channeloffset = 1`. Then add the spectrums together and divide by 2 to get the average spectrum for both channels.

#### What the data represents.

To work out what each entry in the array represents, use this formula

```
entry_hz = output_rate / 2 / numvalues
```

The array represents amplitudes of each frequency band from 0hz to the nyquist rate. The nyquist rate is equal to the output rate divided by 2.

For example when FMOD is set to 44100hz output, the range of represented frequencies will be 0hz to 22049hz, a total of 22050hz represented.

If in the same example, 1024 was passed to this function as the numvalues, each entry's contribution would be as follows.

```
entry_hz = (44100 / 2 / 1024
```

```
entry_hz = 21.53 hz
```

**Note:** This function only displays data for sounds playing that were created with [FMOD\\_SOFTWARE](#). [FMOD\\_HARDWARE](#) based sounds are played using the sound card driver and are not accessible.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_DSP\\_FFT\\_WINDOW](#)
- [System::getSpectrum](#)
- [Channel::getSpectrum](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)

# ChannelGroup::getSystemObject

Retrieves the parent System object that created this channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::getSystemObject(
    FMOD_SYSTEM **system
);
```

## Parameters

*system*

Address of a variable that receives the System object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::createChannelGroup](#)
- [\\_System::getMasterChannelGroup](#)

# ChannelGroup::getUserData

Retrieves the user value that that was set by calling the [ChannelGroup::setUserData](#) function.?

## Syntax

```
FMOD_RESULT ChannelGroup::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the to user data specified with the [ChannelGroup::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::setUserData](#)

# ChannelGroup::getVolume

Retrieves the master volume level for the channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::getVolume (
    float * volume
);
```

## Parameters

*volume*

Address of a variable to receive the channel group volume level, from 0.0 to 1.0 inclusive. 0.0 = silent, 1.0 = full volume. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::setVolume](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)

# ChannelGroup::getWaveData

Retrieves a pointer to a block of PCM data that represents the currently playing waveform for this channel group.  
?This function is useful for a very easy way to plot an oscilloscope.?

## Syntax

```
FMOD_RESULT ChannelGroup::getWaveData (
    float * wavearray,
    int numvalues,
    int channeloffset
);
```

## Parameters

*wavearray*

Address of a variable that receives the currently playing waveform data. This is an array of floating point values.

*numvalues*

Number of floats to write to the array. Maximum value = 16384.

*channeloffset*

Offset into multichannel data. Mono channels use 0. Stereo channels use 0 = left, 1 = right. More than stereo use the appropriate index.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This is the actual resampled, filtered and volume scaled data, at the time this function is called.

Do not use this function to try and display the whole waveform of the sound, as this is more of a 'snapshot' of the current waveform at the time it is called, and could return the same data if it is called very quickly in succession. See the DSP API to capture a continual stream of wave data as it plays, or see [Sound::lock](#) / [Sound::unlock](#) if you want to simply display the waveform of a sound.

This function allows retrieval of left and right data for a stereo sound individually. To combine them into one signal, simply add the entries of each separate buffer together and then divide them by 2.

**Note:** This function only displays data for sounds playing that were created with [FMOD\\_SOFTWARE](#). [FMOD\\_HARDWARE](#) based sounds are played using the sound card driver and are not accessible.



## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::getMasterChannelGroup](#)
- [\\_System::createChannelGroup](#)
- [\\_Sound::lock](#)
- [\\_Sound::unlock](#)

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# ChannelGroup::override3DAttributes

Overrides the position and velocity of all channels within this channel group and those of any sub channelgroups.?

## Syntax

```
FMOD_RESULT ChannelGroup::override3DAttributes (
    const FMOD_VECTOR * pos ,
    const FMOD_VECTOR * vel
);
```

## Parameters

*pos*

Position in 3D space of the channels in the group. Specifying 0 / null will ignore this parameter.

*vel*

Velocity in 'distance units per second' in 3D space of the group of channels. See remarks. Specifying 0 / null will ignore this parameter.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A 'distance unit' is specified by [System::set3DSettings](#). By default this is set to meters which is a distance scale of 1.0.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::set3DAttributes](#)
- [Channel::get3DAttributes](#)
- [FMOD\\_VECTOR](#)
- [System::set3DSettings](#)



# ChannelGroup::overrideFrequency

Overrides the frequency or playback rate, in HZ of all channels within this channel group and those of any sub channelgroups.?

## Syntax

```
FMOD_RESULT ChannelGroup::overrideFrequency (
    float frequency
);
```

## Parameters

*frequency*

A frequency value in HZ. This value can also be negative to play the sound backwards (negative frequencies allowed with [FMOD\\_SOFTWARE](#) based non-stream sounds only). DirectSound hardware voices have limited frequency range on some soundcards. Please see remarks for more on this.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When a sound is played, it plays at the default frequency of the sound which can be set by Sound::setDefault. For most file formats, the volume is determined by the audio format.

Frequency limitations for sounds created with [FMOD\\_HARDWARE](#) in DirectSound.

Every hardware device has a minimum and maximum frequency. This means setting the frequency above the maximum and below the minimum will have no effect.

FMOD clamps frequencies to these values when playing back on hardware, so if you are setting the frequency outside of this range, the frequency will stay at either the minimum or maximum.

Note that [FMOD\\_SOFTWARE](#) based sounds do not have this limitation.

To find out the minimum and maximum value before initializing FMOD (maybe to decide whether to use a different soundcard, output mode, or drop back fully to software mixing), you can use the [System::getDriverCaps](#) function.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Channel::setFrequency](#)
- [\\_Channel::getFrequency](#)
- [\\_System::getDriverCaps](#)
- [\\_System::getMasterChannelGroup](#)
- [\\_System::createChannelGroup](#)

# ChannelGroup::overridePan

Sets pan position linearly of all channels within this channel group and those of any sub channelgroups.?

## Syntax

```
FMOD_RESULT ChannelGroup::overridePan(
    float pan
);
```

## Parameters

*pan*

A left/right pan level, from -1.0 to 1.0 inclusive. -1.0 = Full left, 0.0 = center, 1.0 = full right. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Panning only works on sounds created with [FMOD\\_2D](#). 3D sounds are not pannable.

Only sounds that are mono or stereo can be panned. Multichannel sounds (ie >2 channels) cannot be panned.

Mono sounds are panned from left to right using constant power panning. This means when pan = 0.0, the balance for the sound in each speaker is 71% left and 71% right, not 50% left and 50% right. This gives (audibly) smoother pans. Stereo sounds have each left/right value faded up and down according to the specified pan position. This means when pan = 0.0, the balance for the sound in each speaker is 100% left and 100% right. When pan = -1.0, only the left channel of the stereo sound is audible, when pan = 1.0, only the right channel of the stereo sound is audible.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)
- [Channel::setPan](#)
- [Channel::getPan](#)



# ChannelGroup::overrideReverbProperties

Overrides the reverb properties of all channels within this channel group and those of any sub channelgroups.?

## Syntax

```
FMOD_RESULT ChannelGroup::overrideReverbProperties (
    const FMOD_REVERB_CHANNELPROPERTIES * prop
);
```

## Parameters

*prop*

Pointer to a [FMOD\\_REVERB\\_CHANNELPROPERTIES](#) structure definition.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

With [FMOD\\_HARDWARE](#) on Windows using EAX, the reverb will only work on [FMOD\\_3D](#) based sounds. [FMOD\\_SOFTWARE](#) does not have this problem and works on [FMOD\\_2D](#) and [FMOD\\_3D](#) based sounds.

On PlayStation 2, the 'Room' parameter is the only parameter supported. The hardware only allows 'on' or 'off', so the reverb will be off when 'Room' is -10000 and on for every other value.

On Xbox, it is possible to apply reverb to [FMOD\\_2D](#) and [FMOD\\_HARDWARE](#) based voices using this function. By default reverb is turned off for [FMOD\\_2D](#) hardware based voices, to make it compatible with EAX.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_REVERB\\_CHANNELPROPERTIES](#)
- [System::setReverbProperties](#)
- [System::getReverbProperties](#)



- [\\_Channel::setReverbProperties](#)
- [\\_Channel::getReverbProperties](#)
- [\\_System::getMasterChannelGroup](#)
- [\\_System::createChannelGroup](#)

Version 4.12.03 Built on Feb 18, 2008

# ChannelGroup::overrideSpeakerMix

Overrides all channel speaker levels for each speaker individually.?

## Syntax

```
MOD_RESULT channelGroupOverrideSpeakerMix(  
    float frontLft,  
    float frontRht,  
    float center,  
    float lfe,  
    float backLft,  
    float backRht,  
    float sideLft,  
    float sideRht  
);
```

## Parameters

*frontleft*

Level for this channel in the front left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

*frontright*

Level for this channel in the front right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

*center*

Level for this channel in the center speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

*lfe*

Level for this channel in the subwoofer speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

*backleft*

Level for this channel in the back left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

*backright*

Level for this channel in the back right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

*sideleft*

Level for this channel in the side left speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

*sideright*

Level for this channel in the side right speaker of a multichannel speaker setup. 0.0 = silent, 1.0 = full volume.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function only works on sounds created with [FMOD\\_2D](#). 3D sounds are not pannable and will return [FMOD\\_ERR\\_NEEDS2D](#).

Only sounds create with [FMOD\\_SOFTWARE](#) playing on this channel will allow this functionality.

Speakers specified that don't exist will simply be ignored.

For more advanced speaker control, including sending the different channels of a stereo sound to arbitrary speakers, see [Channel::setSpeakerLevels](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setSpeakerMix](#)
- [Channel::getSpeakerMix](#)
- [Channel::setSpeakerLevels](#)

# ChannelGroup::overrideVolume

Overrides the volume of all channels within this channel group and those of any sub channelgroups.?

## Syntax

```
FMOD_RESULT ChannelGroup::overrideVolume (
    float volume
);
```

## Parameters

*volume*

A linear volume level, from 0.0 to 1.0 inclusive. 0.0 = silent, 1.0 = full volume. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

This is not to be used as a master volume for the group, as it will modify the volumes of the channels themselves.

If you want to scale the volume of the group, use [ChannelGroup::setVolume](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)
- [ChannelGroup::setVolume](#)

# ChannelGroup::release

Frees a channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

All channels assigned to this group are returned back to the master channel group owned by the System object. See [System::getMasterChannelGroup](#).

All child groups assigned to this group are returned back to the master channel group owned by the System object. See [System::getMasterChannelGroup](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createChannelGroup](#)
- [System::getMasterChannelGroup](#)

# ChannelGroup::set3DOcclusion

Sets the master occlusion factors for the channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::set3DOcclusion(
    float directOcclusion,
    float reverbOcclusion
);
```

## Parameters

*directOcclusion*

Occlusion factor for the direct path. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0.

*reverbOcclusion*

Occlusion factor for the reverb mix. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function does not go through and overwrite the channel occlusion factors. It scales them by the channel group's occlusion factors.

That way when [Channel::set3DOcclusion](#) / [Channel::get3DOcclusion](#) is called the respective individual channel occlusion factors will still be preserved. This means that final Channel occlusion values will be affected by both ChannelGroup occlusion and geometry (if any).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::get3DOcclusion](#)
- [Channel::set3DOcclusion](#)
- [Channel::get3DOcclusion](#)
- [System::getMasterChannelGroup](#)

Version 4.12.03 Built on Feb 18, 2008

# ChannelGroup::setMute

Mutes a channelgroup, and the channels within it, or unmutes any unmuted channels if set to false.?

## Syntax

```
FMOD_RESULT ChannelGroup::setMute (
    bool mute
);
```

## Parameters

*mute*

Mute state to set. true = channelgroup state is set to muted. false = channelgroup state is set to unmuted.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A channelgroup maintains a mute state, that affects channelgroups and channels within it. If a channelgroup is muted, all channelgroups and channels below it will become muted.

Channels will not have their per channel mute state overwritten, so that when a channelgroup is unmuted, the muted state of the channels will correct as they were set on a per channel basis.

This means even though a channel is muted, it can return false when you call [Channel::getMute](#) on that channel, because that was the state of the channel at the time before the ChannelGroup was muted.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getMute](#)
- [Channel::setMute](#)
- [Channel::getMute](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)





# ChannelGroup::setPaused

Pauses a channelgroup, and the channels within it, or unpauses any unpaused channels if set to false.?

## Syntax

```
FMOD_RESULT ChannelGroup::setPaused(  
    bool paused  
);
```

## Parameters

*paused*

Paused state to set. true = channelgroup state is set to paused. false = channelgroup state is set to unpaused.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A channelgroup maintains a paused state, that affects channelgroups and channels within it. If a channelgroup is paused, all channelgroups and channels below it will become paused.

Channels will not have their per channel pause state overwritten, so that when a channelgroup is unpaused, the paused state of the channels will correct as they were set on a per channel basis.

This means even though a channel is paused, it can return false when you call [Channel::getPaused](#) on that channel, because that was the state of the channel at the time before the ChannelGroup was paused.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getPaused](#)
- [Channel::setPaused](#)
- [Channel::getPaused](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)



# ChannelGroup::setPitch

Sets the master pitch for the channel group.?

## Syntax

```
FMOD_RESULT ChannelGroup::setPitch(
    float pitch
);
```

## Parameters

*pitch*

A pitch level, from 0.0 to 10.0 inclusive. 0.5 = half pitch, 2.0 = double pitch. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function does not go through and overwrite the channel frequencies. It scales them by the channel group's pitch. That way when [Channel::setFrequency](#) / [Channel::getFrequency](#) is called the respective individual channel frequencies will still be preserved.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::overrideFrequency](#)
- [System::getMasterChannelGroup](#)
- [ChannelGroup::getPitch](#)
- [Channel::setFrequency](#)
- [Channel::getFrequency](#)
- [ChannelGroup::overrideFrequency](#)

# ChannelGroup::setUserData

Sets a user value that the ChannelGroup object will store internally. Can be retrieved with [ChannelGroup::getUserData](#).

## Syntax

```
FMOD_RESULT ChannelGroup::setUserData (
    void * userData
);
```

## Parameters

*userData*

Address of user data that the user wishes stored within the ChannelGroup object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [ChannelGroup::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::getUserData](#)
- [System::getMasterChannelGroup](#)
- [System::createChannelGroup](#)

# ChannelGroup::setVolume

Sets the master volume for the channel group linearly.?

## Syntax

```
FMOD_RESULT ChannelGroup::setVolume (
    float volume
);
```

## Parameters

*volume*

A linear volume level, from 0.0 to 1.0 inclusive. 0.0 = silent, 1.0 = full volume. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function does not go through and overwrite the channel volumes. It scales them by the channel group's volume. That way when [Channel::setVolume](#) / [Channel::getVolume](#) is called the respective individual channel volumes will still be preserved.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [ChannelGroup::setVolume](#)
- [Channel::setVolume](#)
- [Channel::getVolume](#)
- [ChannelGroup::overrideVolume](#)

# ChannelGroup::stop

Stops all channels within the channelgroup.?

## Syntax

```
FMOD_RESULT ChannelGroup::stop();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::playSound](#)
- [\\_System::getMasterChannelGroup](#)
- [\\_System::createChannelGroup](#)

# SoundGroup Interface

[\\_SoundGroup::getMaxAudible](#)  
[SoundGroup::getMaxAudibleBehavior](#)  
[SoundGroup::getMuteFadeSpeed](#)  
[SoundGroup::getName](#)  
[SoundGroup::getNumPlaying](#)  
[SoundGroup::getNumSounds](#)  
[SoundGroup::getSound](#)  
[SoundGroup::getSystemObject](#)  
[SoundGroup::getUserData](#)  
[SoundGroup::getVolume](#)  
[SoundGroup::release](#)  
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[SoundGroup::setMuteFadeSpeed](#)  
[SoundGroup::setUserData](#)  
[SoundGroup::setVolume](#)  
[SoundGroup::stop](#)



# SoundGroup::getMaxAudible

Retrieves the number of concurrent playbacks of sounds in a sound group to the specified value.  
If the sounds in the sound group are playing this many times, any attempts to play more of the sounds in the sound group will fail with [FMOD\\_ERR\\_MAXAUDIBLE](#).

## Syntax

```
FMOD_RESULT SoundGroup::getMaxAudible (
    int * maxaudible
);
```

## Parameters

*maxaudible*

Address of a variable to receive the number of playbacks to be audible at once. -1 = unlimited. 0 means no sounds in this group will succeed. Default = -1.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

[SoundGroup::getNumPlaying](#) can be used to determine how many instances of the sounds in the sound group are playing.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::setMaxAudible](#)
- [SoundGroup::getNumPlaying](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::getMaxAudibleBehavior

Retrieves the current max audible behavior method.?

## Syntax

```
FMOD_RESULT SoundGroup::getMaxAudibleBehavior(  
    FMOD_SOUNDGROUP_BEHAVIOR * behavior  
);
```

## Parameters

*behavior*

Address of a variable to receive the current sound group max playbacks behavior. Default is [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_FAIL](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_SOUNDGROUP\\_BEHAVIOR](#)
- [SoundGroup::setMaxAudibleBehavior](#)
- [SoundGroup::setMaxAudible](#)
- [SoundGroup::getMaxAudible](#)
- [SoundGroup::setMuteFadeSpeed](#)
- [SoundGroup::getMuteFadeSpeed](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::getMuteFadeSpeed

Retrieves the current time in seconds for [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_MUTE](#) behavior to fade with.?

## Syntax

```
FMOD_RESULT SoundGroup::getMuteFadeSpeed(
    float * speed
);
```

## Parameters

*speed*

Address of a variable to receive the fade time in seconds (1.0 = 1 second). Default = 0.0. (no fade).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a mode besides [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_MUTE](#) is used, the fade speed is ignored.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::setMuteFadeSpeed](#)
- [SoundGroup::setMaxAudibleBehavior](#)
- [SoundGroup::getMaxAudibleBehavior](#)
- [SoundGroup::setMaxAudible](#)
- [SoundGroup::getMaxAudible](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::getName

Retrieves the name of the sound group.?

## Syntax

```
FMOD_RESULT SoundGroup::getName (
    char * name,
    int nameLen
);
```

## Parameters

*name*

Address of a variable that receives the name of the sound group.

*namelen*

Length in bytes of the target buffer to receive the string.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::getNumPlaying

Retrieves the number of currently playing channels for the sound group.?

## Syntax

```
FMOD_RESULT SoundGroup::getNumPlaying (
    int * numPlaying
);
```

## Parameters

*numPlaying*

Address of a variable to receive the number of actively playing channels from sounds in this sound group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This routine returns the number of channels playing. If the sound group only has 1 sound, and that sound is playing twice, the figure returned will be 2.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::getNumSounds

Retrieves the current number of sounds in this sound group.?

## Syntax

```
FMOD_RESULT SoundGroup::getNumSounds (
    int *    numsounds
);
```

## Parameters

*numsounds*

Address of a variable to receive the number of sounds in this sound group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)
- [SoundGroup::setMaxAudible](#)
- [SoundGroup::getSound](#)

# SoundGroup::getSound

Retrieves a pointer to a sound from within a sound group.?

## Syntax

```
FMOD_RESULT SoundGroup::getSound(  
    int index,  
    FMOD_Sound ** sound  
);
```

## Parameters

*index*

Index of the sound that is to be retrieved.

*sound*

Address of a variable to receive a pointer to a Sound object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use [SoundGroup::getNumSounds](#) in conjunction with this function to enumerate all sounds in a sound group.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createSoundGroup](#)
- [System::createSound](#)
- [SoundGroup::getNumSounds](#)
- [System::getMasterSoundGroup](#)





# SoundGroup::getSystemObject

Retrieves the parent System object that was used to create this object.?

## Syntax

```
FMOD_RESULT SoundGroup::getSystemObject(
    FMOD_System ** system
);
```

## Parameters

*system*

Address of a pointer that receives the System object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::createSoundGroup](#)
- [\\_System::getMasterSoundGroup](#)

# SoundGroup::getUserData

Retrieves the user value that that was set by calling the [SoundGroup::setUserData](#) function.?

## Syntax

```
FMOD_RESULT SoundGroup::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [SoundGroup::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::setUserData](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::getVolume

Retrieves the volume for the sounds within a soundgroup.?

## Syntax

```
FMOD_RESULT SoundGroup::getVolume (
    float * volume
);
```

## Parameters

*volume*

Address of a variable to receive the soundgroup volume level, from 0.0 to 1.0 inclusive. 0.0 = silent, 1.0 = full volume. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::setVolume](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::release

Releases a soundgroup object and returns all sounds back to the master sound group.?

## Syntax

```
FMOD_RESULT SoundGroup::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

You cannot release the master sound group.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::setMaxAudible

Limits the number of concurrent playbacks of sounds in a sound group to the specified value.

After this, if the sounds in the sound group are playing this many times, any attempts to play more of the sounds in the sound group will by default fail with [FMOD\\_ERR\\_MAXAUDIBLE](#).

Use [SoundGroup::setMaxAudibleBehavior](#) to change the way the sound playback behaves when too many sounds are playing. Muting, failing and stealing behaviors can be specified.

?

## Syntax

```
FMOD_RESULT SoundGroup::setMaxAudible (
    int maxAudible
);
```

## Parameters

*maxaudible*

Number of playbacks to be audible at once. -1 = unlimited. 0 means no sounds in this group will succeed. Default = -1.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

[SoundGroup::getNumPlaying](#) can be used to determine how many instances of the sounds in the sound group are currently playing.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createSoundGroup](#)
- [SoundGroup::getMaxAudible](#)
- [SoundGroup::getNumPlaying](#)
- [SoundGroup::setMaxAudibleBehavior](#)
- [SoundGroup::getMaxAudibleBehavior](#)
- [System::getMasterSoundGroup](#)

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# SoundGroup::setMaxAudibleBehavior

This function changes the way the sound playback behaves when too many sounds are playing in a soundgroup. Muting, failing and stealing behaviors can be specified.  
?

## Syntax

```
FMOD_RESULT SoundGroup::setMaxAudibleBehavior(  
    FMOD_SOUNDGROUP_BEHAVIOR behavior  
);
```

## Parameters

*behavior*

Specify a behavior determined with a [FMOD\\_SOUNDGROUP\\_BEHAVIOR](#) flag. Default is [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_FAIL](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_SOUNDGROUP\\_BEHAVIOR](#)
- [SoundGroup::getMaxAudibleBehavior](#)
- [SoundGroup::setMaxAudible](#)
- [SoundGroup::getMaxAudible](#)
- [SoundGroup::setMuteFadeSpeed](#)
- [SoundGroup::getMuteFadeSpeed](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::setMuteFadeSpeed

Specify a time in seconds for [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_MUTE](#) behavior to fade with. By default there is no fade.

?When more sounds are playing in a SoundGroup than are specified with [SoundGroup::setMaxAudible](#), the least important sound (ie lowest priority / lowest audible volume due to 3d position, volume etc) will fade to silence if [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_MUTE](#) is used, and any previous sounds that were silent because of this rule will fade in if they are more important.

?

## Syntax

```
FMOD_RESULT SoundGroup::setMuteFadeSpeed(
    float speed
);
```

## Parameters

*speed*

Fade time in seconds (1.0 = 1 second). Default = 0.0. (no fade).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a mode besides [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_MUTE](#) is used, the fade speed is ignored.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::getMuteFadeSpeed](#)
- [SoundGroup::setMaxAudibleBehavior](#)
- [SoundGroup::getMaxAudibleBehavior](#)
- [SoundGroup::setMaxAudible](#)
- [SoundGroup::getMaxAudible](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)



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# SoundGroup::setUserData

Sets a user value that the SoundGroup object will store internally. Can be retrieved with [SoundGroup::getUserData](#)?

## Syntax

```
FMOD_RESULT SoundGroup::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the sound group object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [SoundGroup::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::getUserData](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::setVolume

Sets the volume for a sound group, affecting all channels playing the sounds in this soundgroup.?

## Syntax

```
FMOD_RESULT SoundGroup::setVolume (
    float volume
);
```

## Parameters

*volume*

A linear volume level, from 0.0 to 1.0 inclusive. 0.0 = silent, 1.0 = full volume. Default = 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [SoundGroup::getVolume](#)
- [System::createSoundGroup](#)
- [System::getMasterSoundGroup](#)

# SoundGroup::stop

Stops all sounds within this soundgroup.?

## Syntax

```
FMOD_RESULT SoundGroup::stop();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::playSound](#)
- [\\_System::createSoundGroup](#)
- [\\_System::getMasterSoundGroup](#)

# DSP Interface

[DSP::addInput](#)  
[DSP::disconnectAll](#)  
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[DSP::setOutputLevels](#)  
[DSP::setOutputMix](#)  
[DSP::setParameter](#)  
[DSP::setUserData](#)  
[DSP::showConfigDialog](#)

# DSP::addInput

Adds the specified DSP unit as an input of the DSP object.?

## Syntax

```
FMOD_RESULT DSP::addInput(  
    FMOD_DSP * target  
);
```

## Parameters

*target*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration. Adding a unit as an input means that there can be multiple units added to the target.

Inputs are automatically mixed together, then the mixed data is sent to the unit's output(s).

To find the number of inputs or outputs a unit has use [DSP::getNumInputs](#) or [DSP::getNumOutputs](#).

## Remarks

If you want to add a unit as an output of another unit, then add 'this' unit as an input of that unit instead.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getNumInputs](#)
- [DSP::getInput](#)
- [DSP::getNumOutputs](#)
- [DSP::disconnectFrom](#)

# DSP::disconnectAll

Helper function to disconnect either all inputs or all outputs of a dsp unit.?

## Syntax

```
FMOD_RESULT DSP::disconnectAll(  
    bool inputs ,  
    bool outputs  
);
```

## Parameters

*inputs*

*outputs*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is optimized to be faster than disconnecting inputs and outputs manually one by one.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::disconnectFrom](#)

# DSP::disconnectFrom

Disconnect the DSP unit from the specified target.?

## Syntax

```
FMOD_RESULT DSP::disconnectFrom (
    FMOD_DSP * target
);
```

## Parameters

*target*

The unit that this unit is to be removed from. Specify 0 or NULL to disconnect the unit from all outputs and inputs.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note that when you disconnect a unit, it is up to you to reconnect the network so that data flow can continue.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::addInput](#)
- [\\_DSP::disconnectAll](#)



# DSP::getActive

Retrieves the active state of a DSP unit.?

## Syntax

```
FMOD_RESULT DSP::getActive (
    bool * active
);
```

## Parameters

*active*

Address of a variable that receives the active state of the unit. true = unit is activated, false = unit is deactivated.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::setActive](#)
- [DSP::setBypass](#)

# DSP::getBypass

Retrieves the bypass state of the DSP unit.?

## Syntax

```
FMOD_RESULT DSP::getBypass (  
    bool * bypass  
);
```

## Parameters

*bypass*

Address of a variable that receives the bypass state for a DSP unit. true = unit is not processing audio data, false = unit is processing audio data. Default = false.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a unit is bypassed, it will still process its inputs, unlike [DSP::setActive](#) (when set to false) which causes inputs to stop processing as well.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::setBypass](#)
- [DSP::setActive](#)

# DSP::getDefaults

Retrieves the default frequency, volume, pan and more for this DSP unit if it was to ever be played on a channel using [System::playDSP](#)?

## Syntax

```
FMOD_RESULT DSP::getDefaults (
    float * frequency,
    float * volume,
    float * pan,
    int * priority
);
```

## Parameters

*frequency*

Address of a variable that receives the default frequency for the DSP unit. Optional. Specify 0 or NULL to ignore.

*volume*

Address of a variable that receives the default volume for the DSP unit. Result will be from 0.0 to 1.0. 0.0 = Silent, 1.0 = full volume. Default = 1.0. Optional. Specify 0 or NULL to ignore.

*pan*

Address of a variable that receives the default pan for the DSP unit. Result will be from -1.0 to +1.0. -1.0 = Full left, 0.0 = center, 1.0 = full right. Default = 0.0. Optional. Specify 0 or NULL to ignore.

*priority*

Address of a variable that receives the default priority for the DSP unit when played on a channel. Result will be from 0 to 256. 0 = most important, 256 = least important. Default = 128. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::setDefaults](#)

- [\\_System:playDSP](#)

Version 4.12.03 Built on Feb 18, 2008

# DSP::getInfo

Retrieves information about the current DSP unit, including name, version, default channels and width and height of configuration dialog box if it exists.?

## Syntax

```
FMOD_RESULT DSP::getInfo (
    char * name,
    unsigned int * version,
    int * channels,
    int * configwidth,
    int * configheight
);
```

## Parameters

*name*

Address of a variable that receives the name of the unit. This will be a maximum of 32bytes. If the DSP unit has filled all 32 bytes with the name with no terminating \0 null character it is up to the caller to append a null character. Optional. Specify 0 or NULL to ignore.

*version*

Address of a variable that receives the version number of the DSP unit. Version number is usually formatted as hex AAAABBBB where the AAAA is the major version number and the BBBB is the minor version number. Optional. Specify 0 or NULL to ignore.

*channels*

Address of a variable that receives the number of channels the unit was initialized with. 0 means the plugin will process whatever number of channels is currently in the network. >0 would be mostly used if the unit is a unit that only generates sound, or is not flexible enough to take any number of input channels. Optional. Specify 0 or NULL to ignore.

*configwidth*

Address of a variable that receives the width of an optional configuration dialog box that can be displayed with [DSP::showConfigDialog](#). 0 means the dialog is not present. Optional. Specify 0 or NULL to ignore.

*configheight*

Address of a variable that receives the height of an optional configuration dialog box that can be displayed with [DSP::showConfigDialog](#). 0 means the dialog is not present. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::showConfigDialog](#)

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# DSP::getInput

Retrieves a pointer to a DSP unit which is acting as an input to this unit.?

## Syntax

```
FMOD_RESULT DSP::getInput(  
    int index,  
    FMOD_DSP** input  
);
```

## Parameters

*index*

Index of the input unit to retrieve.

*input*

Address of a variable that receives the pointer to the desired input unit.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

An input is a unit which feeds audio data to this unit.

If there are more than 1 input to this unit, the inputs will be mixed, and the current unit processes the mixed result.

Find out the number of input units to this unit by calling [DSP::getNumInputs](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getNumInputs](#)
- [DSP::addInput](#)
- [DSP::getOutput](#)





# DSP::getInputLevels

Retrieves the speaker mix for a DSP unit's input.?

## Syntax

```
FMOD_RESULT DSP::getInputLevels (
    int index,
    FMOD_SPEAKER speaker,
    float * levels,
    int numlevels
);
```

## Parameters

*index*

DSP input index to get the speaker levels from. The number of inputs for a DSP unit can be found with [DSP::getNumInputs](#).

*speaker*

The target speaker to get the levels from. This can be cast to an integer if you are using a device with more than the pre-defined speaker range.

*levels*

Address of an array of floating point numbers to get the speaker levels of an input.

*numlevels*

The number of floats within the levels parameter being passed to this function. In the case of the above mono or stereo sound, 1 or 2 could be used respectively. If the sound being played was an 8 channel multichannel sound then 8 levels would be used.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::setInputLevels](#)

- [DSP::getNumInputs](#)
- [DSP::getOutputLevels](#)

Version 4.12.03 Built on Feb 18, 2008

# DSP::getInputMix

Retrieves the volume of the specified input to be mixed into this unit.?

## Syntax

```
FMOD_RESULT DSP::getInputMix(  
    int index,  
    float * volume  
);
```

## Parameters

*index*

Input index to retrieve the volume for. The number of inputs for a DSP unit can be found with [DSP::getNumInputs](#).

*volume*

Address of a variable to receive the volume or mix level of the specified input. 0.0 = silent, 1.0 = full volume.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::setInputMix](#)
- [DSP::getNumInputs](#)
- [DSP::getOutputMix](#)

# DSP::getNumInputs

Retrieves the number of inputs connected to the DSP unit.?

## Syntax

```
FMOD_RESULT DSP::getNumInputs (
    int * numinputs
);
```

## Parameters

*numinputs*

Address of a variable that receives the number of inputs connected to this unit.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Inputs are units that feed data to this unit. When there are multiple inputs, they are mixed together.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getNumOutputs](#)
- [DSP::getInput](#)

# DSP::getNumOutputs

Retrieves the number of outputs connected to the DSP unit.?

## Syntax

```
FMOD_RESULT DSP::getNumOutputs (
    int * numoutputs
);
```

## Parameters

*numoutputs*

Address of a variable that receives the number of outputs connected to this unit.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Outputs are units that this unit feeds data to. When there are multiple outputs, the data is split and sent to each unit individually.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getNumInputs](#)
- [DSP::getOutput](#)

# DSP::getNumParameters

Retrieves the number of parameters a DSP unit has to control its behaviour.?

## Syntax

```
FMOD_RESULT DSP::getNumParameters (  
  int * numParams  
);
```

## Parameters

*numParams*

Address of a variable that receives the number of parameters contained within this DSP unit.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use this to enumerate all parameters of a DSP unit with [DSP::getParameter](#) and [DSP::getParameterInfo](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [DSP::getParameterInfo](#)

# DSP::getOutput

Retrieves a pointer to a DSP unit which is acting as an output to this unit.?

## Syntax

```
FMO D RESULT S P: ge to u tp t(  
    i nt i nd x,  
    FMO D: : S P ** ou t p t  
);
```

## Parameters

*index*

Index of the output unit to retrieve.

*output*

Address of a variable that receives the pointer to the desired output unit.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

An output is a unit which this unit will feed data too once it has processed its data.

Find out the number of output units to this unit by calling [DSP::getNumOutputs](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getNumOutputs](#)
- [DSP::addInput](#)
- [DSP::getInput](#)





# DSP::getOutputLevels

Retrieves the speaker mix for a DSP unit's output.?

## Syntax

```
FMOD_RESULT DSP::getOutputLevels (
    int index,
    FMOD_SPEAKER speaker,
    float * levels,
    int numlevels
);
```

## Parameters

*index*

DSP output index to get the speaker levels from. The number of outputs for a DSP unit can be found with [DSP::getNumOutputs](#).

*speaker*

The target speaker to get the levels from. This can be cast to an integer if you are using a device with more than the pre-defined speaker range.

*levels*

Address of an array of floating point numbers to get the speaker levels of an output.

*numlevels*

The number of floats within the levels parameter being passed to this function. In the case of the above mono or stereo sound, 1 or 2 could be used respectively. If the sound being played was an 8 channel multichannel sound then 8 levels would be used.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::setOutputLevels](#)

- [DSP::getNumOutputs](#)
- [DSP::getOutputLevels](#)

Version 4.12.03 Built on Feb 18, 2008

# DSP::getOutputMix

Retrieves the volume of the specified output to be mixed into this unit.?

## Syntax

```
FMOD_RESULT DSP::getOutputMix(  
    int index,  
    float * volume  
);
```

## Parameters

*index*

Output index to retrieve the volume for. The number of outputs for a DSP unit can be found with [DSP::getNumOutputs](#).

*volume*

Address of a variable to receive the volume or mix level of the specified output. 0.0 = silent, 1.0 = full volume.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::setOutputMix](#)
- [\\_DSP::getNumOutputs](#)
- [\\_DSP::getInputMix](#)

# DSP::getParameter

Retrieves a DSP unit's parameter by index. To find out the parameter names and range, see the [see also](#) field.?

## Syntax

```
FMOD_RESULT DSP::getParameter(  
    int index,  
    float * value,  
    char * valuestr,  
    int valuestrlen  
);
```

## Parameters

*index*

Parameter index for this unit. Find the number of parameters with [DSP::getNumParameters](#).

*value*

Address of a variable that receives the parameter value. The parameter properties can be retrieved with [DSP::getParameterInfo](#).

*valuestr*

Address of a variable that receives the string containing a formatted or more meaningful representation of the DSP parameter's value. For example if a switch parameter has on and off (0.0 or 1.0) it will display "ON" or "OFF" by using this parameter.

*valuestrlen*

Length of the user supplied memory in bytes that valuestr will write to. This will not exceed 16 bytes.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getParameterInfo](#)
- [DSP::getNumParameters](#)

- [DSP::setParameter](#)

Version 4.12.03 Built on Feb 18, 2008

# DSP::getParameterInfo

Retrieve information about a specified parameter within the DSP unit.?

## Syntax

```
MOD_RESULT DSP::getParameterInfo (
    int index,
    char* name,
    char* label,
    char* description,
    int descriptionlen,
    float min,
    float max
);
```

## Parameters

*index*

Parameter index for this unit. Find the number of parameters with [DSP::getNumParameters](#).

*name*

Address of a variable that receives the name of the parameter. An example is "Gain". This is a maximum string length of 16bytes (append \0 in case the plugin has used all 16 bytes for the string).

*label*

Address of a variable that receives the label of the parameter (ie a parameter type that might go next to the parameter). An example is "dB". This is a maximum string length of 16bytes (append \0 in case the plugin has used all 16 bytes for the string).

*description*

Address of a variable that receives the more descriptive text about the parameter (ie for a tooltip). An example is "Controls the input level for the effect in decibels".

*descriptionlen*

Maximum length of user supplied description string in bytes that FMOD will write to.

*min*

Minimum range of the parameter.

*max*

Maximum range of the parameter.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use [DSP::getNumParameters](#) to find out the number of parameters for this DSP unit.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::setParameter](#)
- [\\_DSP::getParameter](#)
- [\\_DSP::getNumParameters](#)

# DSP::getSystemObject

Retrieves the parent System object that was used to create this object.?

## Syntax

```
FMOD_RESULT DSP::getSystemObject(
    FMOD_SYSTEM **system
);
```

## Parameters

*system*

Address of a variable that receives the System object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::createDSP](#)
- [\\_System::createDSPByType](#)
- [\\_System::getDSPHead](#)
- [\\_Channel::getDSPHead](#)
- [\\_ChannelGroup::getDSPHead](#)



# DSP::getType

Retrieves the pre-defined type of a FMOD registered DSP unit.?

## Syntax

```
FMOD_RESULT DSP::getType (
    FMOD_DSP_TYPE * type
);
```

## Parameters

*type*

Address of a variable to receive the FMOD dsp type.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This is only valid for built in FMOD effects. Any user plugins will simply return [FMOD\\_DSP\\_TYPE\\_UNKNOWN](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_FMOD\\_DSP\\_TYPE](#)

# DSP::getUserData

Retrieves the user value that that was set by calling the [DSP::setUserData](#) function.?

## Syntax

```
FMOD_RESULT DSP::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the to user data specified with the [DSP::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::setUserData](#)

# DSP::release

Frees a DSP object.?

## Syntax

```
FMOD_RESULT DSP::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This will free the DSP object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System::createDSP](#)
- [\\_System::createDSPByType](#)
- [\\_System::getDSPHead](#)
- [\\_Channel::getDSPHead](#)
- [\\_ChannelGroup::getDSPHead](#)

# DSP::remove

Removes a unit from a DSP chain and connects the unit's input and output together after it is gone.?

## Syntax

```
FMOD_RESULT DSP::remove(0);
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is generally only used with units that have been added with [System::addDSP](#) or [Channel::addDSP](#).

A unit that has been added in this way generally only has one input and one output, so this function assumes this and takes input 0 and connects it with output 0 after it has been removed, so that the data flow is not broken.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::addDSP](#)
- [Channel::addDSP](#)
- [ChannelGroup::addDSP](#)

# DSP::reset

Calls the DSP unit's reset function, which will clear internal buffers and reset the unit back to an initial state.?

## Syntax

```
FMOD_RESULT DSP::reset();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Calling this function is useful if the DSP unit relies on a history to process itself (ie an echo filter).

If you disconnected the unit and reconnected it to a different part of the network with a different sound, you would want to call this to reset the units state (ie clear and reset the echo filter) so that you dont get left over artifacts from the place it used to be connected.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

# DSP::setActive

Enables or disables a unit for being processed.?

## Syntax

```
FMOD_RESULT DSP::setActive(  
    bool active  
);
```

## Parameters

*active*

true = unit is activated, false = unit is deactivated.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This does not connect or disconnect a unit in any way, it just disables it so that it is not processed.

If a unit is disabled, and has inputs, they will also cease to be processed.

To disable a unit but allow the inputs of the unit to continue being processed, use [DSP::setBypass](#) instead.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getActive](#)
- [DSP::setBypass](#)

# DSP::setBypass

Enables or disables the read callback of a DSP unit so that it does or doesn't process the data coming into it.  
?A DSP unit that is disabled still processes its inputs, it will just be 'dry'.?

## Syntax

```
FMOD_RESULT DSP::setBypass (  
    bool bypass  
);
```

## Parameters

*bypass*

Boolean to cause the read callback of the DSP unit to be bypassed or not. Default = false.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If a unit is bypassed, it will still process its inputs.

To disable the unit and all of its inputs, use [DSP::setActive](#) instead.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getBypass](#)
- [DSP::setActive](#)

# DSP::setDefault

If a DSP unit is to be played on a channel with [System::playDSP](#), this will set the defaults for frequency, volume, pan and more for the channel.

## Syntax

```
FMOD_RESULT DSP::setDefault(  
    float frequency,  
    float volume,  
    float pan,  
    int priority  
);
```

## Parameters

### *frequency*

Default playback frequency for the DSP unit, in hz. (ie 44100hz).

### *volume*

Default volume for the DSP unit. 0.0 to 1.0. 0.0 = Silent, 1.0 = full volume. Default = 1.0.

### *pan*

Default pan for the DSP unit. -1.0 to +1.0. -1.0 = Full left, 0.0 = center, 1.0 = full right. Default = 0.0.

### *priority*

Default priority for the DSP unit when played on a channel. 0 to 256. 0 = most important, 256 = least important. Default = 128.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

There are no 'ignore' values for these parameters. Use [DSP::getDefaults](#) if you want to change only 1 and leave others unaltered.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris



**See Also**

- [System:playDSP](#)
- [DSP:getDefaults](#)

Version 4.12.03 Built on Feb 18, 2008

# DSP::setInputLevels

Sets the speaker mix for a DSP unit's input.?

## Syntax

```
FMOD_RESULT DSP::setInputLevels (
    int index,
    FMOD_SPEAKER speaker,
    float * levels,
    int numlevels
);
```

## Parameters

*index*

DSP input index to set the speaker levels for. The number of inputs for a DSP unit can be found with [DSP::getNumInputs](#).

*speaker*

The target speaker to modify the levels for. This can be cast to an integer if you are using a device with more than the pre-defined speaker range.

*levels*

An array of floating point numbers from 0.0 to 1.0 representing the volume of each input channel of a sound. See remarks for more.

*numlevels*

The number of floats within the levels parameter being passed to this function. In the case of the above mono or stereo sound, 1 or 2 could be used respectively. If the sound being played was an 8 channel multichannel sound then 8 levels would be used.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

As an example of usage of this function, if the sound played on this speaker was mono, only 1 level would be needed.

If the sound played on this channel was stereo, then an array of 2 floats could be specified. For example { 0, 1 } on a channel playing a stereo sound would mute the left part of the stereo sound when it is played on this speaker.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::getInputLevels](#)
- [\\_DSP::getNumInputs](#)
- [\\_DSP::setOutputLevels](#)

Version 4.12.03 Built on Feb 18, 2008

# DSP::setInputMix

Sets the volume of the specified input to be mixed into this unit.?

## Syntax

```
FMOD_RESULT DSP::setInputMix(  
    int index,  
    float volume  
);
```

## Parameters

*index*

Input index to set the volume level for. The number of inputs for a DSP unit can be found with [DSP::getNumInputs](#).

*volume*

Volume or mix level of the specified input. 0.0 = silent, 1.0 = full volume.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getInputMix](#)
- [DSP::getNumInputs](#)
- [DSP::setOutputMix](#)

# DSP::setOutputLevels

Sets the speaker mix for a DSP unit's output.?

## Syntax

```
FMOD_RESULT DSP::setOutputLevels (
    int index,
    FMOD_SPEAKER speaker,
    float * levels,
    int numlevels
);
```

## Parameters

*index*

DSP output index to set the speaker levels for. The number of outputs for a DSP unit can be found with [DSP::getNumOutputs](#).

*speaker*

The target speaker to modify the levels for. This can be cast to an integer if you are using a device with more than the pre-defined speaker range.

*levels*

An array of floating point numbers from 0.0 to 1.0 representing the volume of each output channel of a sound. See remarks for more.

*numlevels*

The number of floats within the levels parameter being passed to this function. In the case of the above mono or stereo sound, 1 or 2 could be used respectively. If the sound being played was an 8 channel multichannel sound then 8 levels would be used.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

As an example of usage of this function, if the sound played on this speaker was mono, only 1 level would be needed.

If the sound played on this channel was stereo, then an array of 2 floats could be specified. For example { 0, 1 } on a channel playing a stereo sound would mute the left part of the stereo sound when it is played on this speaker.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::getOutputLevels](#)
- [\\_DSP::getNumOutputs](#)
- [\\_DSP::setInputLevels](#)

Version 4.12.03 Built on Feb 18, 2008

# DSP::setOutputMix

Sets the volume of the specified output to be mixed into this unit.?

## Syntax

```
FMOD_RESULT DSP::setOutputMix(  
    int index,  
    float volume  
);
```

## Parameters

*index*

Output index to set the volume level for. The number of outputs for a DSP unit can be found with [DSP::getNumOutputs](#).

*volume*

Volume or mix level of the specified output. 0.0 = silent, 1.0 = full volume.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_DSP::getOutputMix](#)
- [\\_DSP::getNumOutputs](#)
- [\\_DSP::setInputMix](#)

# DSP::setParameter

Sets a DSP unit's parameter by index. To find out the parameter names and range, see the [see also](#) field.?

## Syntax

```
FMOD_RESULT DSP::setParameter(  
    int index,  
    float value  
);
```

## Parameters

*index*

Parameter index for this unit. Find the number of parameters with [DSP::getNumParameters](#).

*value*

Parameter value. The parameter properties can be retrieved with [DSP::getParameterInfo](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getParameterInfo](#)
- [DSP::getNumParameters](#)
- [DSP::getParameter](#)



# DSP::setUserData

Sets a user value that the DSP object will store internally. Can be retrieved with [DSP::getUserData](#).

## Syntax

```
FMOD_RESULT DSP::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the DSP object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [DSP::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getUserData](#)

# DSP::showConfigDialog

Display or hide a DSP unit configuration dialog box inside the target window.?

## Syntax

```
FMOD_RESULT DSP::showConfigDialog (
    HWND * hnd,
    bool show
);
```

## Parameters

*hwnd*

Target HWND in windows to display configuration dialog.

*show*

true = show dialog box inside target hwnd. false = remove dialog from target hwnd.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Dialog boxes are used by DSP plugins that prefer to use a graphical user interface to modify their parameters rather than using the other method of enumerating the parameters and using [DSP::setParameter](#).

These are usually VST plugins. FMOD Ex plugins do not have configuration dialog boxes. To find out what size window to create to store the configuration screen, use [DSP::getInfo](#) where you can get the width and height.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [DSP::getInfo](#)
- [DSP::setParameter](#)
- [DSP::getParameter](#)



# Geometry Interface

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[Geometry::setRotation](#)

[Geometry::setScale](#)

[Geometry::setUserData](#)

# Geometry::addPolygon

Adds a polygon to an existing geometry object.?

## Syntax

```
MO_D_E_S_U_I_T_G_e_o_m_e_t_r_y : a_d_d_P_o_l_y_g_o_n(  
    f_b_a_t_d_i_r_e_c_t_o_c_c_l_u_s_i_o_n,  
    f_b_a_t_r_e_v_e_r_b_o_c_c_l_u_s_i_o_n,  
    b_o_o_l_d_o_u_b_l_e_s_i_d_e_d,  
    i_n_t_n_u_m_v_e_r_t_i_c_e_s ,  
    c_o_n_s_t_M_O_D_E_C_T_R_*_v_e_r_t_i_c_e_s ,  
    i_n_t_*_p_o_l_y_g_o_n_i_n_d_e_x  
);
```

## Parameters

### *directocclusion*

Occlusion value from 0.0 to 1.0 which affects volume or audible frequencies. 0.0 = The polygon does not occlude volume or audible frequencies (sound will be fully audible), 1.0 = The polygon fully occludes (sound will be silent).

### *reverbocclusion*

Occlusion value from 0.0 to 1.0 which affects the reverb mix. 0.0 = The polygon does not occlude reverb (reverb reflections still travel through this polygon), 1.0 = The polygon fully occludes reverb (reverb reflections will be silent through this polygon).

### *doublesided*

Description of polygon if it is double sided or single sided. true = polygon is double sided, false = polygon is single sided, and the winding of the polygon (which determines the polygon's normal) determines which side of the polygon will cause occlusion.

### *numvertices*

Number of vertices in this polygon. This must be at least 3. Polygons (more than 3 sides) are supported.

### *vertices*

A pointer to an array of vertices located in object space, with the count being the number of vertices described using the numvertices parameter.

### *polygonindex*

Address of a variable to receive the polygon index for this object. This index can be used later with other per polygon based geometry functions.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### Note!

- All vertices must lay in the same plane otherwise behaviour may be unpredictable.
- The polygon is assumed to be convex. A non convex polygon will produce unpredictable behaviour.
- Polygons with zero area will be ignored.

Vertices of an object are in object space, not world space, and so are relative to the position, or center of the object.

See [Geometry::setPosition](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

### See Also

- [Geometry::getNumPolygons](#)
- [Geometry::setPosition](#)
- [FMOD\\_VECTOR](#)

# Geometry::getActive

Retrieves the user set active state of the geometry object.?

## Syntax

```
FMOD_RESULT Geometry::getActive (
    bool * active
);
```

## Parameters

*active*

Address of a variable to receive the active state of the object. true = active, false = not active. Default = true.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Geometry::setActive](#)

# Geometry::getMaxPolygons

Retrieves the maximum number of polygons and vertices allocatable for this object. This is not the number of polygons or vertices currently present.

?The maximum number was set with [System::createGeometry](#)?

## Syntax

```
FMOD_RESULT Geometry::getMaxPolygons (
    int * maxpolygons ,
    int * maxvertices
);
```

## Parameters

*maxpolygons*

Address of a variable to receive the maximum possible number of polygons in this object.

*maxvertices*

Address of a variable to receive the maximum possible number of vertices in this object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createGeometry](#)
- [System::loadGeometry](#)



# Geometry::getNumPolygons

Retrieves the number of polygons stored within this geometry object.?

## Syntax

```
FMOD_RESULTGeometry::getNumPolygons(  
    int *    numPolygons  
);
```

## Parameters

*numpolygons*

Address of a variable to receive the number of polygons within this object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Polygons are added to a geometry object via Geometry::addPolygon.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::AddPolygon](#)

# Geometry::getPolygonAttributes

Retrieves the attributes for a particular polygon inside a geometry object.?

## Syntax

```
FMOD_RESULT Geometry::getPolygonAttributes (
    int index,
    float * directocclusion,
    float * reverbocclusion,
    bool * doublesided
);
```

## Parameters

*index*

Polygon index inside the object.

*directocclusion*

Address of a variable to receive the occlusion value from 0.0 to 1.0 which affects volume or audible frequencies. 0.0 = The polygon does not occlude volume or audible frequencies (sound will be fully audible), 1.0 = The polygon fully occludes (sound will be silent).

*reverbocclusion*

Address of a variable to receive the occlusion value from 0.0 to 1.0 which affects the reverb mix. 0.0 = The polygon does not occlude reverb (reverb reflections still travel through this polygon), 1.0 = The polygon fully occludes reverb (reverb reflections will be silent through this polygon).

*doublesided*

Address of a variable to receive the description of polygon if it is double sided or single sided. true = polygon is double sided, false = polygon is single sided, and the winding of the polygon (which determines the polygon's normal) determines which side of the polygon will cause occlusion.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Geometry::getPolygonAttributes](#)
- [\\_Geometry::getNumPolygons](#)

# Geometry::getPolygonNumVertices

Gets the number of vertices in a polygon which is part of the geometry object.?

## Syntax

```
FMOD_RESULTGeometry::getPolygonNumVertices (  
    int index,  
    int * numvertices  
);
```

## Parameters

*index*

Polygon index. This must be in the range of 0 to [Geometry::getNumPolygons](#) minus 1.

*numvertices*

Address of a variable to receive the number of vertices for the selected polygon.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getNumPolygons](#)

# Geometry::getPolygonVertex

Retrieves the position of the vertex inside a geometry object.?

## Syntax

```
FMOD_RESULT Geometry::getPolygonVertex(  
    int index,  
    int vertexindex,  
    FMOD_VECTOR * vertex  
);
```

## Parameters

*index*

Polygon index. This must be in the range of 0 to [Geometry::getNumPolygons](#) minus 1.

*vertexindex*

Vertex index inside the polygon. This must be in the range of 0 to [Geometry::getPolygonNumVertices](#) minus 1.

*vertex*

Address of an [FMOD\\_VECTOR](#) structure which will receive the new vertex location in object space.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Vertices are relative to the position of the object. See [Geometry::setPosition](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getPolygonNumVertices](#)
- [Geometry::setPosition](#)
- [Geometry::getNumPolygons](#)
- [FMOD\\_VECTOR](#)

Version 4.12.03 Built on Feb 18, 2008

# Geometry::getPosition

Retrieves the position of the object in 3D world space.?

## Syntax

```
FMOD_RESULT Geometry::getPosition(  
    FMOD_VECTOR * position  
);
```

## Parameters

*position*

Address of a variable to receive the 3d position of the object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::setPosition](#)
- [FMOD\\_VECTOR](#)

# Geometry::getRotation

Retrieves the orientation of the geometry object.?

## Syntax

```
FMOD_RESULT Geometry::getRotation(  
    FMOD_VECTOR * forward,  
    FMOD_VECTOR * up  
);
```

## Parameters

*forward*

Address of a variable that receives the forwards orientation of the geometry object. Specify 0 or NULL to ignore.

*up*

Address of a variable that receives the upwards orientation of the geometry object. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

See remarks in [System::set3DListenerAttributes](#) for more description on forward and up vectors.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::setRotation](#)
- [System::set3DListenerAttributes](#)
- [FMOD\\_VECTOR](#)



# Geometry::getScale

Retrieves the relative scale vector of the geometry object. An object can be scaled/warped in all 3 dimensions separately using the vector.?

## Syntax

```
FMOD_RESULT Geometry::getScale (
    FMOD_VECTOR * scale
);
```

## Parameters

*scale*

Address of a variable to receive the scale vector of the object. Default = 1.0, 1.0, 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::setScale](#)
- [FMOD\\_VECTOR](#)

# Geometry::getUserData

Retrieves the user value that that was set by calling the [Geometry::setUserData](#) function.?

## Syntax

```
FMOD_RESULT Geometry::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [Geometry::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::setUserData](#)

# Geometry::release

Frees a geometry object and releases its memory.?

## Syntax

```
FMOD_RESULTGeometry::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

# Geometry::save

Saves the geometry object as a serialized binary block, to a user memory buffer. This can then be saved to a file if required and loaded later with [System::loadGeometry](#)?

## Syntax

```
FMOD_RESULT Geometry::save (
    void * data,
    int * datasize
);
```

## Parameters

*data*

Address of a variable to receive the serialized geometry object. Specify 0 or NULL to have the datasize parameter return the size of the memory required for this saved object.

*datasize*

Address of a variable to receive the size in bytes required to save this object when 'data' parameter is 0 or NULL.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

To use this function you will normally need to call it twice. Once to get the size of the data, then again to write the data to your pointer.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::loadGeometry](#)
- [System::createGeometry](#)



# Geometry::setActive

Enables or disables an object from being processed in the geometry engine.?

## Syntax

```
FMOD_RESULT Geometry::setActive(  
    bool active  
);
```

## Parameters

*active*

true = active, false = not active. Default = true.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Geometry::getActive](#)

# Geometry::setPolygonAttributes

Sets individual attributes for each polygon inside a geometry object.?

## Syntax

```
FMOD_RESULT Geometry::setPolygonAttributes (
    int index,
    float directocclusion,
    float reverbocclusion,
    bool doublesided
);
```

## Parameters

*index*

Polygon index inside the object.

*directocclusion*

Occlusion value from 0.0 to 1.0 which affects volume or audible frequencies. 0.0 = The polygon does not occlude volume or audible frequencies (sound will be fully audible), 1.0 = The polygon fully occludes (sound will be silent).

*reverbocclusion*

Occlusion value from 0.0 to 1.0 which affects the reverb mix. 0.0 = The polygon does not occlude reverb (reverb reflections still travel through this polygon), 1.0 = The polygon fully occludes reverb (reverb reflections will be silent through this polygon).

*doublesided*

Description of polygon if it is double sided or single sided. true = polygon is double sided, false = polygon is single sided, and the winding of the polygon (which determines the polygon's normal) determines which side of the polygon will cause occlusion.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getPolygonAttributes](#)
- [Geometry::getNumPolygons](#)

Version 4.12.03 Built on Feb 18, 2008



# Geometry::setPolygonVertex

Alters the position of a polygon's vertex inside a geometry object.?

## Syntax

```
FMOD_RESULT Geometry::setPolygonVertex(  
    int index,  
    int vertexindex,  
    const FMOD_VECTOR * vertex  
);
```

## Parameters

*index*

Polygon index. This must be in the range of 0 to [Geometry::getNumPolygons](#) minus 1.

*vertexindex*

Vertex index inside the polygon. This must be in the range of 0 to [Geometry::getPolygonNumVertices](#) minus 1.

*vertex*

Address of an [FMOD\\_VECTOR](#) which holds the new vertex location.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

**Note!** There may be some significant overhead with this function as it may cause some reconfiguration of internal data structures used to speed up sound-ray testing.

You may get better results if you want to modify your object by using [Geometry::setPosition](#), [Geometry::setScale](#) and [Geometry::setRotation](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getPolygonNumVertices](#)

- [Geometry::getPolygonNumVertices](#)
- [Geometry::setPosition](#)
- [Geometry::setScale](#)
- [Geometry::setRotation](#)
- [Geometry::getNumPolygons](#)
- [FMOD\\_VECTOR](#)

Version 4.12.03 Built on Feb 18, 2008

# Geometry::setPosition

Sets the position of the object in world space, which is the same space FMOD sounds and listeners reside in.

## Syntax

```
FMOD_RESULTGeometry::setPosition(  
    const FMOD_VECTOR * position  
);
```

## Parameters

*position*

Pointer to a vector containing the 3d position of the object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getPosition](#)
- [Geometry::setRotation](#)
- [Geometry::setScale](#)
- [FMOD\\_VECTOR](#)

# Geometry::setRotation

Sets the orientation of the geometry object.?

## Syntax

```
FMOD_RESULT Geometry::setRotation(  
    const FMOD_VECTOR * forward,  
    const FMOD_VECTOR * up  
);
```

## Parameters

*forward*

The forwards orientation of the geometry object. This vector must be of unit length and perpendicular to the up vector. You can specify 0 or NULL to not update the forwards orientation of the geometry object.

*up*

The upwards orientation of the geometry object. This vector must be of unit length and perpendicular to the forwards vector. You can specify 0 or NULL to not update the upwards orientation of the geometry object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

See remarks in [System::set3DListenerAttributes](#) for more description on forward and up vectors.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getRotation](#)
- [System::set3DListenerAttributes](#)
- [FMOD\\_VECTOR](#)



# Geometry::setScale

Sets the relative scale vector of the geometry object. An object can be scaled/warped in all 3 dimensions separately using the vector without having to modify polygon data.?

## Syntax

```
FMOD_RESULTGeometry::setScale(  
    const FMOD_VECTOR * scale  
);
```

## Parameters

*scale*

The scale vector of the object. Default = 1.0, 1.0, 1.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getScale](#)
- [Geometry::setRotation](#)
- [Geometry::setPosition](#)
- [FMOD\\_VECTOR](#)

# Geometry::setUserData

Sets a user value that the Geometry object will store internally. Can be retrieved with [Geometry::getUserData](#)?

## Syntax

```
FMOD_RESULT Geometry::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the Geometry object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [Geometry::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Geometry::getUserData](#)

# Reverb Interface

[Reverb::get3DAttributes](#)

[Reverb::getActive](#)

[Reverb::getProperties](#)

[Reverb::getUserData](#)

[Reverb::release](#)

[Reverb::set3DAttributes](#)

[Reverb::setActive](#)

[Reverb::setProperties](#)

[Reverb::setUserData](#)



# Reverb::get3DAttributes

Retrieves the 3d attributes of a Reverb object.?

## Syntax

```
FMOD_RESULT Reverb::get3DAttributes (
    FMOD_VECTOR * position,
    float * mindistance,
    float * maxdistance
);
```

## Parameters

*position*

Address of a variable that will receive the 3d position of the center of the reverb in 3d space. Default = { 0,0,0 }.

*mindistance*

Address of a variable that will receive the distance from the centerpoint that the reverb will have full effect at. Default = 0.0.

*maxdistance*

Address of a variable that will receive the distance from the centerpoint that the reverb will not have any effect. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 3D reverb object is a sphere having 3D attributes (position, minimum distance, maximum distance) and reverb properties.

The properties and 3D attributes of all reverb objects collectively determine, along with the listener's position, the settings of and input gains into a single 3D reverb DSP.

Please note that this only applies to software channels. When the listener is within the sphere of effect of one or more 3d reverbs, the listener's 3D reverb properties are a weighted combination of such 3d reverbs. When the listener is outside all of the reverbs, the 3D reverb setting is set to the default ambient reverb setting.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

**See Also**

- [Reverb::set3DAttributes](#)
- [System::createReverb](#)

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# Reverb::getActive

Retrieves the active state of the reverb object.?

## Syntax

```
FMOD_RESULT Reverb::getActive (
    bool * active
);
```

## Parameters

*active*

Address of a variable to receive the current active state of the reverb object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Reverb::setActive](#)
- [System::createReverb](#)

# Reverb::getProperties

Retrieves the current reverb environment.?

## Syntax

```
FMOD_RESULT Reverb::getProperties (
    FMOD_REVERB_PROPERTIES * properties
);
```

## Parameters

*properties*

Address of a variable that receives the current reverb environment description.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Reverb::setProperties](#)
- [System::createReverb](#)

# Reverb::getUserData

Retrieves the user value that that was set by calling the [Reverb::setUserData](#) function.?

## Syntax

```
FMOD_RESULT Reverb::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [Reverb::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Reverb::setUserData](#)

# Reverb::release

Releases the memory for a reverb object and makes it inactive.?

## Syntax

```
FMOD_RESULT Reverb::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If no reverb objects are created, the ambient reverb will be the only audible reverb. By default this ambient reverb setting is set to OFF.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::createReverb](#)
- [System::setReverbAmbientProperties](#)

# Reverb::set3DAttributes

Sets the 3d properties of a 'virtual' reverb object.?

## Syntax

```
FMOD_RESULT Reverb::set3DAttributes (
    const FMOD_VECTOR * position,
    float mindistance,
    float maxdistance
);
```

## Parameters

*position*

Pointer to a vector containing the 3d position of the center of the reverb in 3d space. Default = { 0,0,0 }.

*mindistance*

The distance from the centerpoint that the reverb will have full effect at. Default = 0.0.

*maxdistance*

The distance from the centerpoint that the reverb will not have any effect. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 3D reverb object is a sphere having 3D attributes (position, minimum distance, maximum distance) and reverb properties.

The properties and 3D attributes of all reverb objects collectively determine, along with the listener's position, the settings of and input gains into a single 3D reverb DSP.

Please note that this only applies to software channels. When the listener is within the sphere of effect of one or more 3d reverbs, the listener's 3D reverb properties are a weighted combination of such 3d reverbs. When the listener is outside all of the reverbs, the 3D reverb setting is set to the default ambient reverb setting.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Reverb::get3DAttributes](#)
- [System::createReverb](#)



# Reverb::setActive

Disables or enables a reverb object so that it does or does not contribute to the 3d scene.?

## Syntax

```
FMOD_RESULT Reverb::setActive(  
    bool active  
);
```

## Parameters

*active*

true = active, false = not active. Default = true.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Reverb::setActive](#)
- [System::createReverb](#)

# Reverb::setProperties

Sets reverb parameters for the current reverb object.

?Reverb parameters can be set manually, or automatically using the pre-defined presets given in the fmod.h header.?

## Syntax

```
FMOD_RESULT Reverb::setProperties (  
    const FMOD_REVERB_PROPERTIES * properties  
);
```

## Parameters

*properties*

Address of an [FMOD\\_REVERB\\_PROPERTIES](#) structure which defines the attributes for the reverb.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [Reverb::getProperties](#)
- [System::createReverb](#)

# Reverb::setUserData

Sets a user value that the Reverb object will store internally. Can be retrieved with [Reverb::getUserData](#)?

## Syntax

```
FMOD_RESULT Reverb::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the Reverb object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [Reverb::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Reverb::getUserData](#)

# Functions

[Debug\\_GetLevel](#)

[Debug\\_SetLevel](#)

[File\\_GetDiskBusy](#)

[File\\_SetDiskBusy](#)

[Memory\\_GetStats](#)

[Memory\\_Initialize](#)

[System\\_Create](#)

# Debug\_GetLevel

Retrieves the current debug logging level.?

## Syntax

```
FMOD_RESULT Debug_GetLevel(
    FMOD_DEBUG_LEVEL * level
);
```

## Parameters

*level*

Address of a variable to receive current debug level.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This only has an effect with 'logging' versions of FMOD Ex. For example on windows it must be via fmodexL.dll, not fmodex.dll.

On Xbox it would be fmodxboxL.lib not fmodxbox.lib.

FMOD\_ERR\_UNSUPPORTED will be returned on non logging versions of FMOD Ex (ie full release).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Debug\\_GetLevel](#)
- [FMOD\\_DEBUGLEVEL](#)

# Debug\_SetLevel

Sets the level of debug logging to the tty / output for logging versions of FMOD Ex.?

## Syntax

```
FMOD_RESULT Debug_SetLevel(
    FMOD_DEBUG_LEVEL level
);
```

## Parameters

*level*

Logging level to set.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This only has an effect with 'logging' versions of FMOD Ex. For example on windows it must be via fmodexL.dll, not fmodex.dll.

On Xbox it would be fmodxboxL.lib not fmodxbox.lib.

FMOD\_ERR\_UNSUPPORTED will be returned on non logging versions of FMOD Ex (ie full release).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Debug\\_GetLevel](#)
- [FMOD\\_DEBUGLEVEL](#)

# File\_GetDiskBusy

Callback for opening a file.?

## Syntax

```
FMOD_RESULT File_GetDiskBusy (  
    int * busy  
);
```

## Parameters

*busy*

Address of an integer to receive the busy state of the disk at the current time.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Do not use this function to synchronize your own reads with, as due to timing, you might call this function and it says false = it is not busy, but the split second after call this function, internally FMOD might set it to busy. Use [File\\_SetDiskBusy](#) for proper mutual exclusion as it uses semaphores.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [File\\_SetDiskBusy](#)

# File\_SetDiskBusy

Mutex function to synchronize user file reads with FMOD's file reads. This function tells fmod that you are using the disk so that it will block until you are finished with it.  
?This function also blocks if FMOD is already using the disk, so that you cannot do a read at the same time FMOD is reading.?

## Syntax

```
FMOD_RESULT File_SetDiskBusy (  
    int busy  
);
```

## Parameters

*busy*

1 = you are about to perform a disk access. 0 = you are finished with the disk.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use this function as a wrapper around your own file reading functions if you want to do simultaneous file reading while FMOD is also reading. ie

```
FMOD_RESULT File_SetDiskBusy (1);  
my_read(...);  
FMOD_RESULT File_SetDiskBusy (0);
```

Warning! This is a critical section internally. If you do not match your busy = true with a busy = false your program may hang!

If you forget to set diskbusy to false it will stop FMOD from reading from the disk.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [File\\_GetDiskBusy](#)





# Memory\_GetStats

Returns information on the memory usage of FMOD. This is useful for determining a fixed memory size to make FMOD work within for fixed memory machines such as consoles.?

## Syntax

```
FMOD_RESULT Memory_GetStats (  
    int * currentAllocated,  
    int * maxAllocated  
);
```

## Parameters

*currentAllocated*

Address of a variable that receives the currently allocated memory at time of call. Optional. Specify 0 or NULL to ignore.

*maxAllocated*

Address of a variable that receives the maximum allocated memory since [System::init](#) or [Memory\\_Initialize](#). Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note that if using FMOD:[Memory\\_Initialize](#), the memory usage will be slightly higher than without it, as FMOD has to have a small amount of memory overhead to manage the available memory.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::init](#)
- [Memory\\_Initialize](#)



# Memory\_Initialize

Specifies a method for FMOD to allocate memory, either through callbacks or its own internal memory management. You can also supply a pool of memory for FMOD to work with and it will do so with no extra calls to malloc or free. This is useful for systems that want FMOD to use their own memory management, or fixed memory devices such as Xbox, Xbox360, PS2 and GameCube that don't want any allocations occurring out of their control causing fragmentation or unpredictable overflows in a tight memory space.

See remarks for more useful information.

?

## Syntax

```
FMOD_RESULT Memory_Initialize(
    void * poolmem,
    int poollen,
    FMOD_MEMORY_CALLBACK use_malloc,
    FMOD_MEMORY_CALLBACK use_realloc,
    FMOD_MEMORY_CALLBACK use_free
);
```

## Parameters

*poolmem*

If you want a fixed block of memory for FMOD to use, pass it in here. Specify the length in poollen. Specifying NULL doesn't use internal management and it relies on callbacks.

*poollen*

Length in bytes of the pool of memory for FMOD to use specified in. Specifying 0 turns off internal memory management and relies purely on callbacks. Length must be a multiple of 512.

*useralloc*

Only supported if pool is NULL. Otherwise it overrides the FMOD internal calls to alloc. Compatible with ansi malloc().

*userrealloc*

Only supported if pool is NULL. Otherwise it overrides the FMOD internal calls to realloc. Compatible with ansi realloc().

*userfree*

Only supported if pool is NULL. Otherwise it overrides the FMOD internal calls to free. Compatible with ansi free().

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

FMOD has been tested to stay in a limit and fail gracefully if the fixed pool size is not large enough with [FMOD\\_ERR\\_MEMORY](#) errors.

FMOD only does allocation when creating streams, music or samples and the FMOD\_Init stage. It never allocates or deallocates memory during the course of runtime processing.

To find out the required fixed size the user can call FMOD: [Memory\\_GetStats](#) with a larger than necessary pool size (or no pool), and find out the maximum ram usage at any one time within FMOD.

FMOD behaves differently based on what you pass into this function in 3 different combinations.

Here are the examples.

```
FMOD:Memory_InitiaI_e(NU LL, 0, NU LL, NU LL, NU LL; // E l b h c k p e y t o
a s i C m a l l o c , e a l l o c a n d f r e e .
FMOD:Memory_InitiaI_e(NU LL, 0, m y a l l o c , m y e a l l o c , m y f r e e); // C a l l b u s e r s u p p l i e d
c a l l b a c k s e e y t i m e F M O D d e s a m e m o y a l l o c a t i o n o r d e a l l o c a t i o n .
FMOD:Memory_InitiaI_e(ptr, l e n, NU LL, NU LL, NU LL; // U s e s " p t r " a n d
m a g e s m e m o y i n t e r a l l y . N e x t a m a l l o c s o r f r e e s a e p e r f o r m e d f r o m t h i s p i n t .
```

Callbacks and memory pools cannot be combined, as if a pool is specified FMOD, manipulates the pool of memory internally with its own allocate and free scheme.

The memory management algorithm to work within a fixed size of ram is extremely efficient and faster than the standard C malloc or free.

On Xbox 1 you MUST specify a pointer and length. The memory provided must be enough to store all sample data.

**NOTE!** Your memory callbacks must be thread safe. If not unexpected behaviour may occur. FMOD calls memory allocation functions from asynchronous threads, such as the thread related to [FMOD\\_NONBLOCKING](#) flag, and sometimes from the mixer thread.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MEMORY\\_ALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_REALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_FREECALLBACK](#)
- [Memory\\_GetStats](#)
- [System::close](#)

# System\_Create

FMOD System creation function. This must be called to create an FMOD System object before you can do anything else. Use this function to create 1, or multiple instances of FMOD System objects.

## Syntax

```
FMOD_RESULT System_Create (
    FMOD_SYSTEM ** sys_ptr
);
```

## Parameters

*system*

Address of a pointer that receives the new FMOD System object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use [System:release](#) to free a system object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System:init](#)
- [System:release](#)

# Callbacks

[FMOD\\_3D\\_ROLLOFFCALLBACK](#)  
[FMOD\\_CHANNEL\\_CALLBACK](#)  
[FMOD\\_CODEC\\_CLOSECALLBACK](#)  
[FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)  
[FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)  
[FMOD\\_CODEC\\_METADATAACALLBACK](#)  
[FMOD\\_CODEC\\_OPENCALLBACK](#)  
[FMOD\\_CODEC\\_READCALLBACK](#)  
[FMOD\\_CODEC\\_SETPOSITIONCALLBACK](#)  
[FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)  
[FMOD\\_DSP\\_CREATECALLBACK](#)  
[FMOD\\_DSP\\_DIALOGCALLBACK](#)  
[FMOD\\_DSP\\_GETPARAMCALLBACK](#)  
[FMOD\\_DSP\\_READCALLBACK](#)  
[FMOD\\_DSP\\_RELEASECALLBACK](#)  
[FMOD\\_DSP\\_RESETCALLBACK](#)  
[FMOD\\_DSP\\_SETPARAMCALLBACK](#)  
[FMOD\\_DSP\\_SETPOSITIONCALLBACK](#)  
[FMOD\\_FILE\\_CLOSECALLBACK](#)  
[FMOD\\_FILE\\_OPENCALLBACK](#)  
[FMOD\\_FILE\\_READCALLBACK](#)  
[FMOD\\_FILE\\_SEEKCALLBACK](#)  
[FMOD\\_MEMORY\\_ALLOCCALLBACK](#)  
[FMOD\\_MEMORY\\_FREECALLBACK](#)  
[FMOD\\_MEMORY\\_REALLOCCALLBACK](#)  
[FMOD\\_OUTPUT\\_CLOSECALLBACK](#)  
[FMOD\\_OUTPUT\\_GETDRIVERCAPSCALLBACK](#)  
[FMOD\\_OUTPUT\\_GETDRIVERNAMECALLBACK](#)  
[FMOD\\_OUTPUT\\_GETHANDLECALLBACK](#)  
[FMOD\\_OUTPUT\\_GETNUMDRIVERSCALLBACK](#)  
[FMOD\\_OUTPUT\\_GETPOSITIONCALLBACK](#)  
[FMOD\\_OUTPUT\\_INITCALLBACK](#)  
[FMOD\\_OUTPUT\\_LOCKCALLBACK](#)  
[FMOD\\_OUTPUT\\_READFROMMIXER](#)  
[FMOD\\_OUTPUT\\_UNLOCKCALLBACK](#)  
[FMOD\\_OUTPUT\\_UPDATECALLBACK](#)  
[FMOD\\_SOUND\\_NONBLOCKCALLBACK](#)  
[FMOD\\_SOUND\\_PCMREADCALLBACK](#)  
[FMOD\\_SOUND\\_PCMSETPOSCALLBACK](#)  
[FMOD\\_SYSTEM\\_CALLBACK](#)

# FMOD\_3D\_ROLLOFFCALLBACK

Callback for system wide 3d channel volume calculation which overrides fmod's internal calculation code.?

## Syntax

```
fba t FCA LLACK MO D3 D R LD FEA LLACK (  
    MO DC A NE L * c h n e l,  
    fba t d s t a n e  
);
```

## Parameters

*channel*

Pointer to a channel handle.

*distance*

Distance in units (meters by default).

## Return Values

## Remarks

C++ Users. Cast **FMOD\_CHANNEL \*** to **FMOD::Channel \*** inside the callback and use as normal.

NOTE: When using the event system, call [Channel::getUserData](#) to get the event instance handle of the event that spawned the channel in question.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::set3DRolloffCallback](#)
- [System::set3DListenerAttributes](#)
- [System::get3DListenerAttributes](#)
- [Channel::getUserData](#)





# FMOD\_CHANNEL\_CALLBACK

Callback for channel events.?

## Syntax

```
FMOD_RESULT FMOD_Channel_SetCallback (
    FMOD_Channel * channel,
    FMOD_Channel_Callback type,
    int command,
    unsigned int commanddata1,
    unsigned int commanddata2
);
```

## Parameters

*channel*

Pointer to a channel handle.

*type*

The type of callback. Refer to [FMOD\\_CHANNEL\\_CALLBACKTYPE](#).

*command*

The command value passed into [Channel::setCallback](#).

*commanddata1*

The first callback type specific data generated by the callback. See remarks for meaning.

*commanddata2*

The second callback type specific data generated by the callback. See remarks for meaning.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

C++ Users. Cast **FMOD\_CHANNEL \*** to **FMOD::Channel \*** inside the callback and use as normal.

'commanddata1' and 'commanddata2' meanings.

These 2 values are set by the callback depending on what is happening in the callback and the type of callback.

- **FMOD\_CHANNEL\_CALLBACKTYPE\_END**

*commanddata1*: Always 0.

*commanddata2*: Always 0.

- **FMOD\_CHANNEL\_CALLBACKTYPE\_VIRTUALVOICE**

*commanddata1*: **0** when voice is swapped from emulated to real. **1** when voice is swapped from real to emulated.

*commanddata2*: Always 0.

- **FMOD\_CHANNEL\_CALLBACKTYPE\_SYNCPOINT**

*commanddata1*: The index of the sync point. Use `Sound::getSyncPointInfo` to retrieve the sync point's attributes.

*commanddata2*: Always 0.

**Note!** Currently the user must call [System::update](#) for these callbacks to trigger!

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Channel::setCallback](#)
- [FMOD\\_CHANNEL\\_CALLBACKTYPE](#)
- [\\_System::update](#)

# FMOD\_CODEC\_CLOSECALLBACK

Close callback for the codec for when FMOD tries to close a sound using this codec.

?This is the callback any codec related memory is freed, and things are generally de-initialized / shut down for the codec.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_CLOSECALLBACK (
    FMOD_CODEC_STATE * codec_state
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_OPENCALLBACK](#)
- [FMOD\\_CODEC\\_READCALLBACK](#)
- [FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)
- [FMOD\\_CODEC\\_SETPositionCALLBACK](#)

- [FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_CODEC\_GETLENGTH\_CALLBACK

Callback to return the length of the song in whatever format required when Sound::getLength is called.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_GETLENGTH_CALLBACK (
    FMOD_CODEC_STATE * codec_state,
    unsigned int * length,
    FMOD_TIMEUNIT lengthtype
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*length*

Address of a variable that is to receive the length of the sound determined by the format specified in the lengthtype parameter.

*lengthtype*

Timeunit type of length to return. This will be one of the timeunits supplied by the codec author in the [FMOD\\_CODEC\\_DESCRIPTION](#) structure.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_TIMEUNIT](#)
- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_OPENCALLBACK](#)
- [FMOD\\_CODEC\\_CLOSECALLBACK](#)
- [FMOD\\_CODEC\\_READCALLBACK](#)
- [FMOD\\_CODEC\\_SETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)

# FMOD\_CODEC\_GETPOSITION\_CALLBACK

Tell callback for the codec for when FMOD tries to get the current position within the with [Channel::getPosition](#)?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_GETPOSITION_CALLBACK (
    FMOD_CODEC_STATE * codec_state,
    unsigned int * position,
    FMOD_TIMEUNIT postype
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*position*

Address of a variable to receive the current position in the codec based on the timeunit specified in the postype parameter.

*postype*

Timeunit type of the position parameter that is requested. This will be one of the timeunits supplied by the codec author in the [FMOD\\_CODEC\\_DESCRIPTION](#) structure.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3



## See Also

- [Channel::getPosition](#)
- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_OPENCALLBACK](#)
- [FMOD\\_CODEC\\_CLOSECALLBACK](#)
- [FMOD\\_CODEC\\_READCALLBACK](#)
- [FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)
- [FMOD\\_CODEC\\_SETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)

# FMOD\_CODEC\_METADATA\_CALLBACK

Callback for sounds that have their?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_METADATA_CALLBACK (
    FMOD_CODEC_STATE * codec_state,
    FMOD_TAG_TYPE type,
    char * name,
    void * data,
    unsigned int datalen,
    FMOD_TAG_DATA_TYPE datatype,
    int unique
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*type*

Source of tag being updated, ie id3v2 or oggvorbis tag for example. See [FMOD\\_TAGDATATYPE](#).

*name*

Name of the tag being updated.

*data*

Contents of tag.

*datalen*

Length of the tag data in bytes.

*datatype*

Data type of tag. Binary / string / unicode etc. See [FMOD\\_TAGDATATYPE](#).

*unique*

If this is true, then the tag (determined by the name) being updated is the only one of its type. If it is false then there are multiple versions of this tag with the same name.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This callback is usually called from sounds that can update their metadata / tag info at runtime. Such a sound could be an internet SHOUTcast / Icecast stream for example.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_OPENCALLBACK](#)
- [FMOD\\_CODEC\\_CLOSECALLBACK](#)
- [FMOD\\_CODEC\\_READCALLBACK](#)
- [FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)
- [FMOD\\_CODEC\\_SETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)
- [FMOD\\_TAGDATATYPE](#)

# FMOD\_CODEC\_OPENCALLBACK

Open callback for the codec for when FMOD tries to open a sound using this codec. This is the callback the file format check is done in, codec related memory is allocated, and things are generally initialized / set up for the codec.

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_OPENCALLBACK (
    FMOD_CODEC_STATE * codec_state,
    FMOD_MODE usermode,
    FMOD_CREATESOUNDEXINFO * userexinfo
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*usermode*

Mode that the user supplied via [System::createSound](#). This is informational and can be ignored, or used if it has relevance to your codec.

*userexinfo*

Extra info structure that the user supplied via [System::createSound](#). This is informational and can be ignored, or used if it has relevance to your codec.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The usermode and userexinfo parameters tell the codec what was passed in by the user. Generally these can be ignored, as the file format usually determines the format and frequency of the sound.

If you have a flexible format codec (ie you don't mind what output format your codec writes to), you might want to use the parameter that was passed in by the user to specify the output sound format / frequency. For example if you normally create a codec that is always 32bit floating point, the user might supply 16bit integer to save memory, so you could use this information to decode your data to this format instead of the original default format.

Read and seek within the file using the 'fileread' and 'fileseek' members of the FMOD\_CODEC codec that is passed

in.

Note: **DO NOT USE YOUR OWN FILESYSTEM.**

The reasons for this are:

- The user may have set their own file system via user filesystem callbacks.
- FMOD allows file reading via disk, memory and TCP/IP. If you use your own file routines you will lose this ability.

**Important!** FMOD will ping all codecs trying to find the right one for the file the user has passed in. Make sure the first line of your codec open is a FAST format check. If it reads an identifying string, checks it and returns an error [FMOD\\_ERR\\_FORMAT](#) if it is not found.

There may be a lot of codecs loaded into FMOD, so you don't want yours slowing down the [System::createSound](#) call because it is inefficient in determining if it is the right format or not.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [System::createSound](#)
- [FMOD\\_CREATESOUNDEXINFO](#)
- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_CLOSECALLBACK](#)
- [FMOD\\_CODEC\\_READCALLBACK](#)
- [FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)
- [FMOD\\_CODEC\\_SETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)

# FMOD\_CODEC\_READCALLBACK

Read callback for the codec for when FMOD tries to read some data from the file to the destination format (format specified in the open callback).?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_READCALLBACK (
    FMOD_CODEC_STATE * codec_state,
    void * buffer,
    unsigned int size_bytes,
    unsigned int * bytes_read
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*buffer*

Buffer to read PCM data to. Note that the format of this data is the format described in FMOD\_CODEC\_WAVEFORMAT.

*sizebytes*

Number of bytes to read

*bytesread*

Number of bytes actually read

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If you cannot read number of bytes requested, simply return [FMOD\\_OK](#) and give bytesread the number of bytes you read.

Read and seek within the file using the 'fileread' and 'fileseek' members of the FMOD\_CODEC codec that is passed in.

Note: **DO NOT USE YOUR OWN FILESYSTEM.**

The reasons for this are:

- The user may have set their own file system via user filesystem callbacks.
- FMOD allows file reading via disk, memory and TCP/IP. If you use your own file routines you will lose this ability.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#)

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_OPENCALLBACK](#)
- [FMOD\\_CODEC\\_CLOSECALLBACK](#)
- [FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)
- [FMOD\\_CODEC\\_SETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)

# FMOD\_CODEC\_SETPosition CALLBACK

Seek callback for the codec for when FMOD tries to seek within the file with [Channel::setPosition?](#)

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_SETPosition TO NA L LACK (
    FMOD_CODEC_STATE * codec_state,
    int subsond,
    unsigned int position,
    FMOD_TIMEUNIT postype
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*subsond*

Subsound within which to seek.

*position*

Position to seek to in the sound based on the timeunit specified in the postype parameter.

*postype*

Timeunit type of the position parameter. This will be one of the timeunits supplied by the codec author in the [FMOD\\_CODEC\\_DESCRIPTION](#) structure.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Read and seek within the file using the 'fileread' and 'fileseek' members of the FMOD\_CODEC codec that is passed in.

Note: **DO NOT USE YOUR OWN FILESYSTEM.**

The reasons for this are:

- The user may have set their own file system via user filesystem callbacks.
- FMOD allows file reading via disk, memory and TCP/IP. If you use your own file routines you will lose this



ability.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#)

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [Channel::setPosition](#)
- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_OPENCALLBACK](#)
- [FMOD\\_CODEC\\_CLOSECALLBACK](#)
- [FMOD\\_CODEC\\_READCALLBACK](#)
- [FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)
- [FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_SOUNDCREATECALLBACK](#)

# FMOD\_CODEC\_SOUNDCREATECALLBACK

Sound creation callback for the codec when FMOD finishes creating the sound. Ie so the codec can set more parameters for the related created sound, ie loop points/mode or 3D attributes etc.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_CODEC_SOUND_CREATE_CALLBACK (
    FMOD_CODEC_STATE * codec_state,
    int sub_sound,
    FMOD_SOUND * sound
);
```

## Parameters

*codec\_state*

Pointer to the codec state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*subsound*

Subsound index being created.

*sound*

Pointer to the sound being created.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [System::createSound](#)
- [System::createStream](#)
- [FMOD\\_CODEC\\_STATE](#)
- [FMOD\\_CODEC\\_DESCRIPTION](#)
- [FMOD\\_CODEC\\_OPENCALLBACK](#)
- [FMOD\\_CODEC\\_CLOSECALLBACK](#)
- [FMOD\\_CODEC\\_READCALLBACK](#)
- [FMOD\\_CODEC\\_GETLENGTHCALLBACK](#)
- [FMOD\\_CODEC\\_SETPOSITIONCALLBACK](#)
- [FMOD\\_CODEC\\_GETPOSITIONCALLBACK](#)

# FMOD\_DSP\_CREATECALLBACK

This callback is called once when a user creates a DSP unit of this type. It is used to allocate memory, initialize variables and the like.

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_DSP_CREATECALLBACK (
    FMOD_DSP_STATE * dsp_state
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

[System::createDSP](#)

[System::createDSPByType](#)

[System::createDSPByIndex](#)

Sometimes a user will re-use a DSP unit instead of releasing it and creating a new one, so it may be useful to implement [FMOD\\_DSP\\_RESETCALLBACK](#) to reset any variables or buffers when the user calls it.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_DSP\\_STATE](#)
- [System::createDSP](#)
- [System::createDSPByType](#)
- [System::createDSPByIndex](#)
- [FMOD\\_DSP\\_RESETCALLBACK](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_DSP\_DIALOGCALLBACK

This callback is called when the user wants the plugin to display a configuration dialog box. This is not always necessary, so this can be left blank if wanted.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_DSP_DIALOGCALLBACK (
    FMOD_DSP_STATE * dsp_state,
    int show,
    void * hwnd
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

*show*

1 = show the dialog, 0 = hide/remove the dialog.

*hwnd*

This is the target hwnd to display the dialog in. It must not pop up on this hwnd, it must actually be drawn within it.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

[DSP::showConfigDialog](#).

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

#### See Also

- [FMOD\\_DSP\\_STATE](#)
- [DSP::showConfigDialog](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_DSP\_GETPARAMCALLBACK

This callback is called when the user wants to get an indexed parameter from a DSP unit.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_DSP_GETPARAMCALLBACK (
    FMOD_DSP_STATE * dsp_state,
    int index,
    float * value,
    char * valuestr
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

*index*

The index into the parameter list for the parameter the user wants to get.

*value*

Pointer to a floating point variable to receive the selected parameter value.

*valuestr*

A pointer to a string to receive the value of the selected parameter, but in text form. This might be useful to display words instead of numbers. For example "ON" or "OFF" instead of 1.0 and 0.0. The length of the buffer being passed in is always 16 bytes, so do not exceed this.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

[DSP::getParameter](#).

FMOD\_DSP\_GETPARAMCALLBACK.



Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#)

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_DSP\\_STATE](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_SETPARAMCALLBACK](#)

# FMOD\_DSP\_READCALLBACK

This callback is called back regularly when the unit has been created, inserted to the DSP network, and set to active by the user.

?This callback requires the user to fill the output pointer with data. Incoming data is provided and may be filtered on its way to the output pointer.

?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_DSP_READCALLBACK (
    FMOD_DSP_STATE * dsp_state,
    float * inbuffer,
    float * outbuffer,
    unsigned int length,
    int inchannels,
    int outchannels
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

*inbuffer*

Pointer to incoming floating point -1.0 to +1.0 ranged data.

*outbuffer*

Pointer to outgoing floating point -1.0 to +1.0 ranged data. The dsp writer must write to this pointer else there will be silence.

*length*

The length of the incoming and outgoing buffer in samples. To get the length of the buffer in bytes, the user must multiply this number by the number of channels coming in (and out, they may be different) and then multiply by 4 for 1 float = 4 bytes.

*inchannels*

The number of channels of interleaved PCM data in the inbuffer parameter. A mono signal coming in would be 1. A stereo signal coming in would be 2.

*outchannels*

The number of channels of interleaved PCM data in the outbuffer parameter. A mono signal going out would be 1. A stereo signal going out would be 2.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

*None.*

This callback is called automatically and periodically when the DSP engine updates.

For a read update to be called it would have to be enabled, and this is done with [DSP::setActive](#).

Data passed into the callback is always floating point, and of the range -1.0 to +1.0. This is a soft limit though, because FMOD will clip it to these ranges in the final stage of the pipeline, so the dsp unit writer does not have to worry about this.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_DSP\\_STATE](#)
- [DSP::setActive](#)

# FMOD\_DSP\_RELEASECALLBACK

This callback is called when the user releases the DSP unit. It is used to free any resources allocated during the course of the lifetime of the DSP or perform any shut down code needed to clean up the DSP unit.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_DSP_RELEASECALLBACK (
    FMOD_DSP_STATE * dsp_state
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

[DSP::release](#)

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_DSP\\_STATE](#)
- [DSP::release](#)



# FMOD\_DSP\_RESETCALLBACK

This callback function is called by [DSP::reset](#) to allow the effect to reset itself to a default state. This is useful if an effect is for example still holding audio data for a sound that has stopped, and the unit wants to be relocated to a new sound. Resetting the unit would clear any buffers, put the effect back to its initial state, and get it ready for new sound data.

## Syntax

```
FMOD_RESULT FMOD_DSP_RESETCALLBACK (
    FMOD_DSP_STATE * dsp_state
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

[DSP::reset](#)

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_DSP\\_STATE](#)
- [DSP::reset](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_DSP\_SETPARAMCALLBACK

This callback is called when the user wants to set a parameter for a DSP unit.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_DSP_SETPARAMCALLBACK (
    FMOD_DSP_STATE * dsp_state,
    int index,
    float value
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

*index*

The index into the parameter list for the parameter the user wants to set.

*value*

The value passed in by the user to set for the selected parameter.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

[DSP::setParameter](#).

Range checking is not needed. FMOD will clamp the incoming value to the specified min/max.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported



Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

**See Also**

- [FMOD\\_DSP\\_STATE](#)
- [DSP::setParameter](#)
- [FMOD\\_DSP\\_GETPARAMCALLBACK](#)

# FMOD\_DSP\_SETPOSITIONCALLBACK

Callback that is called when the user sets the position of a channel with [Channel::setPosition](#)?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_DSP_SETPOSITIONCALLBACK (
    FMOD_DSP_STATE * dsp_state,
    unsigned int position
);
```

## Parameters

*dsp\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data. Do not cast this to FMOD\_DSP! The handle to the user created DSP handle is stored within the [FMOD\\_DSP\\_STATE](#) structure.

*position*

Position in channel stream to set to. Units are PCM samples (ie [FMOD\\_TIMEUNIT\\_PCM](#)).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Functions that the user would have to call for this callback to be called.

[Channel::setPosition](#).

If a DSP unit is attached to a channel and the user calls [Channel::setPosition](#) then this function will be called.

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_DSP\\_STATE](#)
- [Channel::setPosition](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_FILE\_CLOSECALLBACK

Callback for closing a file.?

## Syntax

```
FMOD_RESULT FMOD_FILE_CLOSECALLBACK (
    void * handle,
    void * userdata
);
```

## Parameters

*handle*

This is the handle returned from the open callback to use for your own file routines.

*userdata*

Userdata initialized in the [FMOD\\_FILE\\_OPENCALLBACK](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Close any user created file handle and perform any cleanup necessary for the file here. If the callback is from [System::attachFileSystem](#), then the return value is ignored.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [System::setFileSystem](#)
- [System::attachFileSystem](#)
- [FMOD\\_FILE\\_OPENCALLBACK](#)
- [FMOD\\_FILE\\_READCALLBACK](#)
- [FMOD\\_FILE\\_SEEKCALLBACK](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_FILE\_OPENCALLBACK

Callback for opening a file.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_FILE_OPEN_CALLBACK (
    const char * name,
    int unicode,
    unsigned int * filesize,
    void ** handle,
    void ** userdata
);
```

## Parameters

*name*

This is the filename passed in by the user. You may treat this as you like.

*unicode*

Tells the callback if the string being passed in is a double byte unicode string or not. You may have to support this unless you know the target application will not support unicode.

*filesize*

The size of the file to be passed back to fmod, in bytes.

*handle*

This is to store a handle generated by the user. This will be the handle that gets passed into the other callbacks. Optional but may be needed.

*userdata*

This is to store userdata to be passed into the other callbacks. Optional.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Return the appropriate error code such as [FMOD\\_ERR\\_FILE\\_NOTFOUND](#) if the file fails to open. If the callback

is from [System::attachFileSystem](#), then the return value is ignored.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_System::setFileSystem](#)
- [\\_System::attachFileSystem](#)
- [\\_FMODE\\_FILE\\_CLOSECALLBACK](#)
- [\\_FMODE\\_FILE\\_READCALLBACK](#)
- [\\_FMODE\\_FILE\\_SEEKCALLBACK](#)

# FMOD\_FILE\_READCALLBACK

Callback for reading from a file.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_FILE_READCALLBACK (
    void * handle,
    void * buffer,
    unsigned int size_bytes,
    unsigned int * bytes_read,
    void * userdata
);
```

## Parameters

*handle*

This is the handle you returned from the open callback to use for your own file routines.

*buffer*

The buffer to read your data into.

*sizebytes*

The number of bytes to read.

*bytesread*

The number of bytes successfully read.

*userdata*

Userdata initialized in the [FMOD\\_FILE\\_OPENCALLBACK](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If the callback is from [System::attachFileSystem](#), then the return value is ignored.



## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_System::setFileSystem](#)
- [\\_System::attachFileSystem](#)
- [\\_FMODE\\_FILE\\_OPENCALLBACK](#)
- [\\_FMODE\\_FILE\\_CLOSECALLBACK](#)
- [\\_FMODE\\_FILE\\_SEEKCALLBACK](#)

# FMOD\_FILE\_SEEKCALLBACK

Callback for seeking within a file.?

## Syntax

```
FMOD_RESULT FMOD_FILE_SEEKCALLBACK (
    void * handle,
    unsigned int pos,
    void * userdata
);
```

## Parameters

*handle*

This is the handle returned from the open callback to use for your own file routines.

*pos*

This is the position or offset to seek to in the file in bytes.

*userdata*

Data initialized in the [FMOD\\_FILE\\_OPENCALLBACK](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [System::setFileSystem](#)
- [FMOD\\_FILE\\_OPENCALLBACK](#)
- [FMOD\\_FILE\\_CLOSECALLBACK](#)
- [FMOD\\_FILE\\_READCALLBACK](#)



# FMOD\_MEMORY\_ALLOCCALLBACK

Callback to allocate a block of memory.?

## Syntax

```
void * FMOD_MEMORY_ALLOCCALLBACK (
    unsigned int size,
    FMOD_MEMORY_TYPE type
);
```

## Parameters

*size*

Size in bytes of the memory block to be allocated and returned.

*type*

Type of memory allocation.

## Return Values

On success, a pointer to the newly allocated block of memory is returned.

On failure, NULL is returned.

## Remarks

Returning an aligned pointer, of 16 byte alignment is recommended for speed purposes.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Memory\\_Initialize](#)
- [Memory\\_GetStats](#)
- [FMOD\\_MEMORY\\_REALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_FREECALLBACK](#)
- [FMOD\\_MEMORY\\_TYPE](#)



# FMOD\_MEMORY\_FREECALLBACK

Callback to free a block of memory.?

## Syntax

```
void FMOD_MEMORY_FREECALLBACK (
    void * ptr,
    FMOD_MEMORY_TYPE type
);
```

## Parameters

*ptr*

Pointer to a pre-existing block of memory to be freed.

*type*

Type of memory to be freed.

## Return Values

void

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Memory\\_Initialize](#)
- [Memory\\_GetStats](#)
- [FMOD\\_MEMORY\\_ALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_REALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_TYPE](#)

# FMOD\_MEMORY\_REALLOCCALLBACK

Callback to re-allocate a block of memory to a different size.?

## Syntax

```
void * FMOD_MEMORY_REALLOCCALLBACK (
    void * ptr,
    unsigned int size,
    FMOD_MEMORY_TYPE type
);
```

## Parameters

*ptr*

Pointer to a block of memory to be resized. If this is NULL then a new block of memory is simply allocated.

*size*

Size of the memory to be reallocated. The original memory must be preserved.

*type*

Type of memory allocation.

## Return Values

On success, a pointer to the newly re-allocated block of memory is returned.

On failure, NULL is returned.

## Remarks

Returning an aligned pointer, of 16 byte alignment is recommended for speed purposes.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Memory\\_Initialize](#)

- [Memory\\_GetStats](#)
- [FMOD\\_MEMORY\\_ALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_FREECALLBACK](#)
- [FMOD\\_MEMORY\\_TYPE](#)

Version 4.12.03 Built on Feb 18, 2008



# FMOD\_OUTPUT\_CLOSECALLBACK

Shut down callback which is called when the user calls [System::close](#) or [System::release](#). ([System::release](#) calls [System::close](#) internally)?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_CLOSECALLBACK (
    FMOD_OUTPUT_STATE * output_state
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [System::release](#)
- [System::close](#)

# FMOD\_OUTPUT\_GETDRIVERCAPSCALLBACK

Called when the user calls [System::getDriverCaps](#)?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUTGETDRIVERCAPSCALLBACK (
    FMOD_OUTPUTSTATE * output_state,
    int id,
    FMOD_CAPS * caps,
    int * minfrequency,
    int * maxfrequency,
    FMOD_SPEAKERMODE * controlpanel_speaker_mode
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*id*

Index into the total number of outputs possible, provided by the [FMOD\\_OUTPUT\\_GETNUMDRIVERSCALLBACK](#) callback.

*caps*

Address of a variable to receive the caps available by this output device. See [FMOD\\_CAPS](#). Fill this in.

*minfrequency*

*maxfrequency*

*controlpanelspeakermode*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [\\_System::getDriverCaps](#)
- [\\_System::getDriverInfo](#)
- [\\_System::getNumDrivers](#)
- [FMOD\\_OUTPUT\\_GETNUMDRIVERSCALLBACK](#)

# FMOD\_OUTPUT\_GETDRIVERNAMECALLBACK

Called when the user calls [System::getDriverInfo](#)?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_GETDRIVERNAMECALLBACK (
    FMOD_OUTPUT_STATE * output_state,
    int id,
    char * name,
    int namelen
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*id*

Index into the total number of outputs possible, provided by the [FMOD\\_OUTPUT\\_GETNUMDRIVERSCALLBACK](#) callback.

*name*

Address of a variable to receive the driver name relevant to the index passed in. Fill this in.

*namelen*

Length of name buffer being passed in by the user.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

#### **See Also**

- [System:getDriverInfo](#)
- [System:getNumDrivers](#)
- [FMOD\\_OUTPUT\\_GETNUMDRIVERSCALLBACK](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_OUTPUT\_GETHANDLE CALLBACK

Called when the user calls System::getOutputHandle.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_GETHANDLE (
    FMOD_OUTPUT_STATE * output_state,
    void ** handle
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*handle*

Address of a variable to receive the current plugin's output 'handle'. This is only if the plugin writer wants to allow the user access to the main handle behind the plugin (for example the file handle in a file writer plugin). The pointer type must be published to the user somehow, as is done in fmod.h.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3



# FMOD\_OUTPUT\_GETNUMDRIVERSCALLBACK

Called when the user calls [System::getNumDrivers](#)?

## Syntax

```
FMOD_RESULT CALLBACK FMOD_OUTPUTGETNUMDRIVERSCALLBACK (
    FMOD_OUTPUTSTATE * output_state,
    int * numdrivers
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*numdrivers*

Address of a variable to receive the number of output drivers in your plugin.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

Optional. FMOD will assume 0 if this is not specified.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [System::getNumDrivers](#)
- [System::getDriverInfo](#)
- [FMOD\\_OUTPUT\\_GETDRIVERNAMECALLBACK](#)



Version 4.12.03 Built on Feb 18, 2008

# FMOD\_OUTPUT\_GETPOSITIONCALLBACK

Returns the current PCM offset or playback position for the output stream.

?Called from the mixer thread, only when the 'polling' member of [FMOD\\_OUTPUT\\_DESCRIPTION](#) is set to **true**.

?The internal FMOD output thread calls this function periodically to determine if it should ask for a block of audio data or not.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_GETPOSITIONCALLBACK (
    FMOD_OUTPUT_STATE * output_state,
    unsigned int * pcm
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*pcm*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_OUTPUT\\_DESCRIPTION](#)
- [FMOD\\_OUTPUT\\_LOCKCALLBACK](#)
- [FMOD\\_OUTPUT\\_UNLOCKCALLBACK](#)

# FMOD\_OUTPUT\_INITCALLBACK

Initialization callback which is called when the user calls [System::init](#).

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_INITCALLBACK (
    FMOD_OUTPUTSTATE * output_state,
    int selecteddriver,
    FMOD_INITFLAGS flags,
    int * outputrate,
    int outputchannels,
    FMOD_SOUND_FORMAT * outputformat,
    int * pbufferlength,
    void * extra_data
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*selecteddriver*

This is the selected driver id that the user chose from calling [System::setDriver](#).

*flags*

Initialization flags passed in by the user.

*outputrate*

Output rate selected by the user. If not possible, change the rate to the closest match.

*outputchannels*

Output channel count selected by the user. For example 1 = mono output. 2 = stereo output.

*outputformat*

Output format specified by the user. If not possible to support, return [FMOD\\_ERR\\_FORMAT](#).

*dspbufferlength*

Size of the buffer fmod will mix to in one mix update. This value is in PCM samples.

*dspnumbuffers*

Number of buffers fmod will mix to in a circular fashion. Multiply this by dspbufferlength to get the total size of the output sound buffer to allocate.

*extradrivervdata*

Data passed in by the user specific to this driver. May be used for any purpose.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_RESULT](#)
- [System::init](#)
- [System::setDriver](#)

# FMOD\_OUTPUT\_LOCKCALLBACK

Called from the mixer thread, only when the 'polling' member of [FMOD\\_OUTPUT\\_DESCRIPTION](#) is set to true.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_LOCKCALLBACK (
    FMOD_OUTPUT_STATE * output_state,
    unsigned int offset,
    unsigned int length,
    void ** ptr1,
    void ** ptr2,
    unsigned int * len1,
    unsigned int * len2
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*offset*

Offset in *bytes* to the position the caller wants to lock in the sample buffer.

*length*

Number of *bytes* the caller want to lock in the sample buffer.

*ptr1*

Address of a pointer that will point to the first part of the locked data.

*ptr2*

Address of a pointer that will point to the second part of the locked data. This will be null if the data locked hasn't wrapped at the end of the buffer.

*len1*

Length of data in *bytes* that was locked for ptr1

*len2*

Length of data in *bytes* that was locked for ptr2. This will be 0 if the data locked hasn't wrapped at the end of the buffer.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_OUTPUT\\_DESCRIPTION](#)
- [FMOD\\_OUTPUT\\_UNLOCKCALLBACK](#)
- [FMOD\\_OUTPUT\\_GETPOSITIONCALLBACK](#)

# FMOD\_OUTPUT\_READFROM MIXER

Called by the plugin, when the 'polling' member of FMOD\_OUTPUT\_DESCRIPTION is set to false.

Use this function from your own driver irq/timer to read some data from FMOD's DSP engine. All of the resulting output caused by playing sounds and specifying effects by the user will be mixed here and written to the memory provided by the plugin writer.

?

## Syntax

```
FMOD_RESULT FCALLBACK FMOD_OUTPUT_READFROM(
    FMOD_OUTPUT_STATE * output_state,
    void * buffer,
    unsigned int length
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*buffer*

Plugin-writer provided memory for the FMOD Ex mixer to write to.

*length*

Length of the buffer in samples.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3





# FMOD\_OUTPUT\_UNLOCKCALLBACK

Called from the mixer thread, only when the 'polling' member of [FMOD\\_OUTPUT\\_DESCRIPTION](#) is set to true.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_UNLOCKCALLBACK (
    FMOD_OUTPUT_STATE * output_state,
    void * ptr1,
    void * ptr2,
    unsigned int len1,
    unsigned int len2
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

*ptr1*

Pointer to the 1st locked portion of sample data, from Sound::lock.

*ptr2*

Pointer to the 2nd locked portion of sample data, from Sound::lock.

*len1*

Length of data in *bytes* that was locked for ptr1

*len2*

Length of data in *bytes* that was locked for ptr2. This will be 0 if the data locked hasn't wrapped at the end of the buffer.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is normally called after data has been read/written to from Sound::lock. This function will do any post

processing necessary and if needed, send it to sound ram.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## **See Also**

- [FMOD\\_OUTPUT\\_DESCRIPTION](#)
- [FMOD\\_OUTPUT\\_LOCKCALLBACK](#)
- [FMOD\\_OUTPUT\\_GETPOSITIONCALLBACK](#)

# FMOD\_OUTPUT\_UPDATECALLBACK

Called when the user calls System::update.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_OUTPUT_UPDATECALLBACK (
    FMOD_OUTPUTSTATE * outputstate
);
```

## Parameters

*output\_state*

Pointer to the plugin state. The user can use this variable to access runtime plugin specific variables and plugin writer user data.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Remember to return [FMOD\\_OK](#) at the bottom of the function, or an appropriate error code from [FMOD\\_RESULT](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

# FMOD\_SOUND\_NONBLOCKCALLBACK

Callback to be called when a sound has finished loading.?

## Syntax

```
FMOD_RESULT FMOD_SOUND_NONBLOCKCALLBACK (
    FMOD_SOUND * sound,
    FMOD_RESULT result
);
```

## Parameters

*sound*

Pointer to the sound. C++ users see remarks.

*result*

Error code. [FMOD\\_OK](#) if sound was created successfully, or an error code otherwise.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

C++ Users. Cast **FMOD\_SOUND \*** to **FMOD::Sound \*** inside the callback and use as normal.

Return code currently ignored.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [System::createSound](#)
- [FMOD\\_CREATEINDEXINFO](#)



# FMOD\_SOUND\_PCMREADCALLBACK

Used for 2 purposes.

?One use is for user created sounds when [FMOD\\_OPENUSER](#) is specified when creating the sound.

?The other use is to 'piggyback' on FMOD's read functions when opening a normal sound, therefore the callee can read (rip) or even write back new PCM data while FMOD is opening the sound.?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_SOUND_PCM_READ_CALLBACK (
    FMOD_SOUND * sound,
    void * data,
    unsigned int datalen
);
```

## Parameters

*sound*

Pointer to the sound. C++ users see remarks.

*data*

Pointer to raw PCM data that the user can either read or write to.

*datalen*

Length of the data in bytes.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

C++ Users. Cast **FMOD\_SOUND \*** to **FMOD::Sound \*** inside the callback and use as normal.

The format of the sound can be retrieved with [Sound::getFormat](#) from this callback. This will allow the user to determine what type of pointer to use if they are not sure what format the sound is.

If the callback is used for the purpose of 'piggybacking' normal FMOD sound loads, then you do not have to do anything at all, and it can be treated as purely informational. The return value is also ignored.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

### See Also

- [Sound::getFormat](#)
- [FMOD\\_SOUND\\_PCMSETPOSCALLBACK](#)
- [System::createSound](#)
- [System::createStream](#)
- [FMOD\\_CREATESOUNDEXINFO](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_SOUND\_PCMSETPOSCALLBACK

Callback for when the caller calls a seeking function such as Channel::setTime or Channel::setPosition.  
?If the sound is a user created sound, this can be used to seek within the user's resource.  
?

## Syntax

```
FMOD_RESULT F_CALLBACK FMOD_SOUND_PCMSETPOSCALLBACK (
    FMOD_SOUND * sound,
    int subound,
    unsigned int position,
    FMOD_TIMEUNIT postype
);
```

## Parameters

*sound*

Pointer to the sound. C++ users see remarks.

*subound*

In a multi subsound type sound (ie fsb/dls/cdda), this will contain the index into the list of sounds.

*position*

Position to seek to that has been requested. This value will be of format [FMOD\\_TIMEUNIT](#) and must be parsed to determine what it is. Generally [FMOD\\_TIMEUNIT\\_PCM](#) will be the most common format.

*postype*

Position type that the user wanted to seek with. If the sound is a user create sound and the seek type is unsupported return [FMOD\\_ERR\\_FORMAT](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

C++ Users. Cast **FMOD\_SOUND \*** to **FMOD::Sound \*** inside the callback and use as normal.

If the callback is used for the purpose of 'piggybacking' normal FMOD sound loads, then you do not have to do anything at all, and it can be treated as purely informational. The return value is also ignored.



## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [FMOD\\_SOUND\\_PCMREADCALLBACK](#)
- [System::createSound](#)
- [System::createStream](#)
- [FMOD\\_CREATESOUNDEXINFO](#)

# FMOD\_SYSTEM\_CALLBACK

Callback for system events.?

## Syntax

```
FMOD_RESULT FMOD_SYSTEM_CALLBACK (
    FMOD_SYSTEM * system,
    FMOD_SYSTEM_CALLBACK_TYPE type,
    unsigned int commanddata1,
    unsigned int commanddata2
);
```

## Parameters

*system*

Pointer to a system handle.

*type*

The type of callback. Refer to [FMOD\\_SYSTEM\\_CALLBACKTYPE](#).

*commanddata1*

The first callback type specific data generated by the callback. See remarks for meaning.

*commanddata2*

The second callback type specific data generated by the callback. See remarks for meaning.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

C++ Users. Cast **FMOD\_SYSTEM \*** to **FMOD::System \*** inside the callback and use as normal.

'commanddata1' and 'commanddata2' meanings.

These 2 values are set by the callback depending on what is happening in the callback and the type of callback.

- **FMOD\_SYSTEM\_CALLBACKTYPE\_DEVICELISTCHANGED**

*commanddata1*: Always 0.

*commanddata2*: Always 0.

- **FMOD\_SYSTEM\_CALLBACKTYPE\_MEMORYALLOCATIONFAILED**

*commanddata1*: A string (char\*) which represents the file and line number of the allocation.

*commanddata2*: The size (int) of the requested allocation.

**Note!** Currently the user must call [System:update](#) for some of these callbacks to trigger! See [FMOD\\_SYSTEM\\_CALLBACKTYPE](#) for details.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_System:setCallback](#)
- [FMOD\\_SYSTEM\\_CALLBACKTYPE](#)
- [\\_System:update](#)

# Structures

[FMOD\\_ADVANCEDSETTINGS](#)

[FMOD\\_CDTOC](#)

[FMOD\\_CODEC\\_DESCRIPTION](#)

[FMOD\\_CODEC\\_STATE](#)

[FMOD\\_CODEC\\_WAVEFORMAT](#)

[FMOD\\_CREATESOUNDEXINFO](#)

[FMOD\\_DSP\\_DESCRIPTION](#)

[FMOD\\_DSP\\_PARAMETERDESC](#)

[FMOD\\_DSP\\_STATE](#)

[FMOD\\_GUID](#)

[FMOD\\_OUTPUT\\_DESCRIPTION](#)

[FMOD\\_OUTPUT\\_STATE](#)

[FMOD\\_REVERB\\_CHANNELPROPERTIES](#)

[FMOD\\_REVERB\\_PROPERTIES](#)

[FMOD\\_TAG](#)

[FMOD\\_VECTOR](#)

# FMOD\_ADVANCEDSETTINGS

Settings for advanced features like configuring memory and cpu usage for the FMOD\_CREATECOMPRESSED\_SAMPLE feature.?

## Structure

```
typedef struct {
    int cbSize;
    int maxMPEGcodecs;
    int maxADPCMcodecs;
    int maxXMAcodecs;
    int maxPCMcodecs;
    int ASIO_NmChannels;
    char ** ASIOChannelList;
    FMOD_DS_EAKER * ASIOSpeakerList;
    int maxDSERBs;
    float HRTF_Again;
    float HRTF_M_Again;
    float HRTFFreq;
    float LowPassFilterCutoff;
    int enableQueueSize;
    unsigned int defaultBlockDefSize;
} FMOD_ADVANCEDSETTINGS;
```

## Members

*cbSize*

[in] Size of this structure. Use sizeof(FMOD\_ADVANCEDSETTINGS) NOTE: This must be set before calling [System::getAdvancedSettings!](#)

*maxMPEGcodecs*

[in/out] Optional. Specify 0 to ignore. For use with FMOD\_CREATECOMPRESSED\_SAMPLE only. Mpeg codecs consume 29,424 bytes per instance and this number will determine how many mpeg channels can be played simultaneously. Default = 16.

*maxADPCMcodecs*

[in/out] Optional. Specify 0 to ignore. For use with FMOD\_CREATECOMPRESSED\_SAMPLE only. ADPCM codecs consume 2,136 bytes per instance (based on FSB encoded ADPCM block size - see remarks) and this number will determine how many ADPCM channels can be played simultaneously. Default = 32.

*maxXMAcodecs*

[in/out] Optional. Specify 0 to ignore. For use with FMOD\_CREATECOMPRESSED\_SAMPLE only. XMA codecs consume 20,512 bytes per instance and this number will determine how many XMA channels can be played simultaneously. Default = 32.

*maxPCMcodecs*

[in/out] Optional. Specify 0 to ignore. For use with PS3 only. PCM codecs consume 12,672 bytes per instance and this number will determine how many streams and PCM voices can be played simultaneously. Default = 16

#### *ASIONumChannels*

[in/out] Optional. Specify 0 to ignore. Number of channels available on the ASIO device.

#### *ASIOChannelList*

[in/out] Optional. Specify 0 to ignore. Pointer to an array of strings (number of entries defined by ASIONumChannels) with ASIO channel names.

#### *ASIOSpeakerList*

[in/out] Optional. Specify 0 to ignore. Pointer to a list of speakers that the ASIO channels map to. This can be called after [System::init](#) to remap ASIO output.

#### *max3DReverbDSPs*

[in/out] Optional. Specify 0 to ignore. The max number of 3d reverb DSP's in the system.

#### *HRTFMinAngle*

[in/out] Optional. Specify 0 to ignore. For use with FMOD\_INIT\_SOFTWARE\_HRTF. The angle (0-360) of a 3D sound from the listener's forward vector at which the HRTF function begins to have an effect. Default = 180.0.

#### *HRTFMaxAngle*

[in/out] Optional. Specify 0 to ignore. For use with FMOD\_INIT\_SOFTWARE\_HRTF. The angle (0-360) of a 3D sound from the listener's forward vector at which the HRTF function begins to have maximum effect. Default = 360.0.

#### *HRTFFreq*

[in/out] Optional. Specify 0 to ignore. For use with FMOD\_INIT\_SOFTWARE\_HRTF. The cutoff frequency of the HRTF's lowpass filter function when at maximum effect. (i.e. at HRTFMaxAngle). Default = 4000.0.

#### *vol0virtualvol*

[in/out] Optional. Specify 0 to ignore. For use with FMOD\_INIT\_VOL0\_BECOMES\_VIRTUAL. If this flag is used, and the volume is 0.0, then the sound will become virtual. Use this value to raise the threshold to a different point where a sound goes virtual.

#### *eventqueuesize*

[in/out] Optional. Specify 0 to ignore. For use with FMOD Event system only. Specifies the number of slots available for simultaneous non blocking loads. Default = 32.

#### *defaultDecodeBufferSize*

[in/out] Optional. Specify 0 to ignore. For streams. This determines the default size of the double buffer (in milliseconds) that a stream uses. Default = 400ms

### **Remarks**

maxMPEGcodecs / maxADPCMcodecs / maxXMAcodecs will determine the maximum cpu usage of playing realtime samples. Use this to lower potential excess cpu usage and also control memory usage.

maxPCMcodecs is for use with PS3 only. It will determine the maximum number of PCM voices that can be played at once. This includes streams of any format and all sounds created \*without\* the FMOD\_CREATECOMPRESSED\_SAMPLE flag.

Memory will be allocated for codecs 'up front' (during [System::init](#)) if these values are specified as non zero. If any are zero, it allocates memory for the codec whenever a file of the type in question is loaded. So if maxMPEGcodecs is 0 for example, it will allocate memory for the mpeg codecs the first time an mp3 is loaded or an mp3 based .FSB file is loaded.

Due to inefficient encoding techniques on certain .wav based ADPCM files, FMOD can need an extra 29720 bytes per codec. This means for lowest memory consumption. Use FSB as it uses an optimal/small ADPCM block size.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setAdvancedSettings](#)
- [System::getAdvancedSettings](#)
- [System::init](#)
- [FMOD\\_MODE](#)

# FMOD\_CDTOC

Structure describing a CD/DVD table of contents?

## Structure

```
typedef struct {
    int numtracks;
    int min[100];
    int sec[100];
    int frame[100];
} FMOD_CDTOC;
```

## Members

*numtracks*

[out] The number of tracks on the CD

*min*

[out] The start offset of each track in minutes

*sec*

[out] The start offset of each track in seconds

*frame*

[out] The start offset of each track in frames

## Remarks

Members marked with [in] mean the user sets the value before passing it to the function.  
Members marked with [out] mean FMOD sets the value to be used after the function exits.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [\\_Sound::getTag](#)





# FMOD\_CODEC\_DESCRIPTION

When creating a codec, declare one of these and provide the relevant callbacks and name for FMOD to use when it opens and reads a file.

## Structure

```
typedef struct {
    const char * name;
    unsigned int version;
    int defaultasstream;
    FMOD_TIMEUNIT timeunits;
    FMOD_CODEC_OPEN_CALLBACK open;
    FMOD_CODEC_CLOSE_CALLBACK close;
    FMOD_CODEC_READ_CALLBACK read;
    FMOD_CODEC_SETPSITON_CALLBACK setposition;
    FMOD_CODEC_GETPSITON_CALLBACK getposition;
    FMOD_CODEC_SOUNDREAD_CALLBACK soundread;
    FMOD_CODEC_GETWAVEFORMAT_CALLBACK getwaveformat;
} FMOD_CODEC_DESCRIPTION;
```

## Members

*name*

[in] Name of the codec.

*version*

[in] Plugin writer's version number.

*defaultasstream*

[in] Tells FMOD to open the file as a stream when calling System::createSound, and not a static sample. Should normally be 0 (FALSE), because generally the user wants to decode the file into memory when using System::createSound. Mainly used for formats that decode for a very long time, or could use large amounts of memory when decoded. Usually sequenced formats such as mod/s3m/xm/it/midi fall into this category. It is mainly to stop users that don't know what they're doing from getting FMOD\_ERR\_MEMORY returned from createSound when they should have in fact called System::createStream or used FMOD\_CREATESTREAM in System::createSound.

*timeunits*

[in] When setposition codec is called, only these time formats will be passed to the codec. Use bitwise OR to accumulate different types.

*open*

[in] Open callback for the codec for when FMOD tries to open a sound using this codec.

*close*

[in] Close callback for the codec for when FMOD tries to close a sound using this codec.

*read*

[in] Read callback for the codec for when FMOD tries to read some data from the file to the destination format (specified in the open callback).

*getlength*

[in] Callback to return the length of the song in whatever format required when Sound::getLength is called.

*setposition*

[in] Seek callback for the codec for when FMOD tries to seek within the file with Channel::setPosition.

*getposition*

[in] Tell callback for the codec for when FMOD tries to get the current position within the with Channel::getPosition.

*soundcreate*

[in] Sound creation callback for the codec when FMOD finishes creating the sound. (So the codec can set more parameters for the related created sound, ie loop points/mode or 3D attributes etc).

*getwaveformat*

[in] Callback to tell FMOD about the waveformat of a particular subsound. This is to save memory, rather than saving 1000 FMOD\_CODEC\_WAVEFORMAT structures in the codec, the codec might have a more optimal way of storing this information.

## Remarks

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_CODEC\\_STATE](#)



# FMOD\_CODEC\_STATE

Codec plugin structure that is passed into each callback.

?

?Set these numsubsounds and waveformat members when called in FMOD\_CODEC\_OPENCALLBACK to tell fmod what sort of sound to create.

?

?The format, channels and frequency tell FMOD what sort of hardware buffer to create when you initialize your code. So if you wrote an MP3 codec that decoded to stereo 16bit integer PCM, you would specify

FMOD\_SOUND\_FORMAT\_PCM16, and channels would be equal to 2.

?

## Structure

```
typedef struct {
    int numsubsounds;
    FMOD_CODEC_WAVEFORMAT * waveformat;
    void * plugindata;
    void * filehandle;
    unsigned int frequency;
    FMOD_FILE_CALLBACK fileseek;
    FMOD_FILE_CALLBACK fileseek;
    FMOD_CODEC_METADATA_CALLBACK metadata;
} FMOD_CODEC_STATE;
```

## Members

### *numsubsounds*

[in] Number of 'subsounds' in this sound. Anything other than 0 makes it a 'container' format (ie CDDA/DLS/FSB etc which contain 1 or more subsounds). For most normal, single sound codec such as WAV/AIFF/MP3, this should be 0 as they are not a container for subsounds, they are the sound by itself.

### *waveformat*

[in] Pointer to an array of format structures containing information about each sample. Can be 0 or NULL if FMOD\_CODEC\_GETWAVEFORMAT callback is preferred. The number of entries here must equal the number of subsounds defined in the subsound parameter. If numsubsounds = 0 then there should be 1 instance of this structure.

### *plugindata*

[in] Plugin writer created data the codec author wants to attach to this object.

### *filehandle*

[out] This will return an internal FMOD file handle to use with the callbacks provided.

### *filesize*

[out] This will contain the size of the file in bytes.

## *fileread*

[out] This will return a callable FMOD file function to use from codec.

## *fileseek*

[out] This will return a callable FMOD file function to use from codec.

## *metadata*

[out] This will return a callable FMOD metadata function to use from codec.

## **Remarks**

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

An FMOD file might be from disk, memory or internet, however the file may be opened by the user.

'numsubsounds' should be 0 if the file is a normal single sound stream or sound. Examples of this would be .WAV, .WMA, .MP3, .AIFF.

'numsubsounds' should be 1+ if the file is a container format, and does not contain wav data itself. Examples of these types would be CDDA (multiple CD tracks), FSB (contains multiple sounds), DLS (contain instruments).

The arrays of format, channel, frequency, length and blockalign should point to arrays of information based on how many subsounds are in the format. If the number of subsounds is 0 then it should point to 1 of each attribute, the same as if the number of subsounds was 1. If subsounds was 100 for example, each pointer should point to an array of 100 of each attribute.

When a sound has 1 or more subsounds, you must play the individual sounds specified by first obtaining the subsound with [Sound::getSubSound](#).

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## **See Also**

- [FMOD\\_SOUND\\_FORMAT](#)
- [FMOD\\_FILE\\_READCALLBACK](#)
- [FMOD\\_FILE\\_SEEKCALLBACK](#)
- [FMOD\\_CODEC\\_METADATAACALLBACK](#)
- [Sound::getSubSound](#)
- [Sound::getNumSubSounds](#)

# FMOD\_CODEC\_WAVEFORMAT

Set these values marked 'in' to tell fmod what sort of sound to create.  
?The format, channels and frequency tell FMOD what sort of hardware buffer to create when you initialize your code. So if you wrote an MP3 codec that decoded to stereo 16bit integer PCM, you would specify FMOD\_SOUND\_FORMAT\_PCM16, and channels would be equal to 2.  
?Members marked as 'out' are set by fmod. Do not modify these. Simply specify 0 for these values when declaring the structure, FMOD will fill in the values for you after creation with the correct function pointers.  
?

## Structure

```
typedef struct {
    char name[ 255];
    FMOD_SOUND_FORMAT format;
    int channels;
    int frequency;
    unsigned int length;
    unsigned int lengthinms;
    int blockalign;
    int bytespersec;
    int bytesperchan;
    FMOD_MODE mode;
    unsigned int channelmask;
} FMOD_CODEC_WAVEFORMAT;
```

## Members

*name*

[in] Name of sound.

*format*

[in] Format for (decompressed) codec output, ie FMOD\_SOUND\_FORMAT\_PCM8, FMOD\_SOUND\_FORMAT\_PCM16.

*channels*

[in] Number of channels used by codec, ie mono = 1, stereo = 2.

*frequency*

[in] Default frequency in hz of the codec, ie 44100.

*lengthbytes*

[in] Length in bytes of the source data.

### *lengthpcm*

[in] Length in decompressed, PCM samples of the file, ie length in seconds \* frequency. Used for Sound::getLength and for memory allocation of static decompressed sample data.

### *blockalign*

[in] Blockalign in decompressed, PCM samples of the optimal decode chunk size for this format. The codec read callback will be called in multiples of this value.

### *loopstart*

[in] Loopstart in decompressed, PCM samples of file.

### *loopend*

[in] Loopend in decompressed, PCM samples of file.

### *mode*

[in] Mode to determine whether the sound should by default load as looping, non looping, 2d or 3d.

### *channelmask*

[in] Microsoft speaker channel mask, as defined for WAVEFORMATEXTENSIBLE and is found in ksmedia.h. Leave at 0 to play in natural speaker order.

## **Remarks**

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

An FMOD file might be from disk, memory or network, however the file may be opened by the user.

'numsubsounds' should be 0 if the file is a normal single sound stream or sound. Examples of this would be .WAV, .WMA, .MP3, .AIFF.

'numsubsounds' should be 1+ if the file is a container format, and does not contain wav data itself. Examples of these types would be CDDA (multiple CD tracks), FSB (contains multiple sounds), MIDI/MOD/S3M/XM/IT (contain instruments).

The arrays of format, channel, frequency, length and blockalign should point to arrays of information based on how many subsounds are in the format. If the number of subsounds is 0 then it should point to 1 of each attribute, the same as if the number of subsounds was 1. If subsounds was 100 for example, each pointer should point to an array of 100 of each attribute.

When a sound has 1 or more subsounds, you must play the individual sounds specified by first obtaining the subsound with [Sound::getSubSound](#).

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii



**See Also**

- [FMOD\\_SOUND\\_FORMAT](#)
- [FMOD\\_FILE\\_READCALLBACK](#)
- [FMOD\\_FILE\\_SEEKCALLBACK](#)
- [FMOD\\_CODEC\\_METADATAACALLBACK](#)
- [Sound::getSubSound](#)
- [Sound::getNumSubSounds](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_CREATE\_SOUND\_EX INFO

Use this structure with [System::createSound](#) when more control is needed over loading.

The possible reasons to use this with [System::createSound](#) are:

?

- Loading a file from memory.?
- Loading a file from within another larger (possibly wad/pak) file, by giving the loader an offset and length.?
- To create a user created / non file based sound.?
- To specify a starting subsound to seek to within a multi-sample sounds (ie FSB/DLS/SF2) when created as a stream.?
- To specify which subsounds to load for multi-sample sounds (ie FSB/DLS/SF2) so that memory is saved and only a subset is actually loaded/read from disk.?
- To specify 'piggyback' read and seek callbacks for capture of sound data as fmod reads and decodes it. Useful for ripping decoded PCM data from sounds as they are loaded / played.?
- To specify a MIDI DLS/SF2 sample set file to load when opening a MIDI file. See below on what members to fill for each of the above types of sound you want to create.?

## Structure

```

typedef struct {
    int cbSize;
    unsigned int length;
    unsigned int frequency;
    int numChannels;
    int defaultFrequency;
    FMOD_SOUND_FORMAT format;
    unsigned int decodedBufferSize;
    int inFinalBlock;
    int numSubBlocks;
    int *initialSamples;
    int initialSamplesTotal;
    FMOD_SOUND_CALLBACK readData;
    FMOD_SOUND_CALLBACK seekData;
    FMOD_SOUND_CALLBACK seekData;
    const char *displayName;
    const char *anyProperty;
    int maxPolyphony;
    void *userData;
    FMOD_SOUND_TYPE suggestedSoundType;
    FMOD_FILE_OPEN_CALLBACK useOpen;
    FMOD_FILE_CLOSE_CALLBACK useClose;
    FMOD_FILE_READ_CALLBACK useRead;
    FMOD_FILE_SEEK_CALLBACK useSeek;
    FMOD_SOUND_CREATE_CALLBACK useCreate;
    FMOD_SOUND_UPDATE_CALLBACK useUpdate;
    unsigned int inFinalBlockPosition;
    FMOD_TIMEUNIT inFinalBlockPositionType;
} FMOD_CREATE_SOUND_EX_INFO;

```

## Members

*cbsize*

[in] Size of this structure. This is used so the structure can be expanded in the future and still work on older versions of FMOD Ex.

*length*

[in] Optional. Specify 0 to ignore. Size in bytes of file to load, or sound to create (in this case only if FMOD\_OPENUSER is used). Required if loading from memory. If 0 is specified, then it will use the size of the file (unless loading from memory then an error will be returned).

*fileoffset*

[in] Optional. Specify 0 to ignore. Offset from start of the file to start loading from. This is useful for loading files from inside big data files.

*numchannels*

[in] Optional. Specify 0 to ignore. Number of channels in a sound mandatory if FMOD\_OPENUSER or FMOD\_OPENRAW is used.

*defaultfrequency*

[in] Optional. Specify 0 to ignore. Default frequency of sound in a sound mandatory if FMOD\_OPENUSER or FMOD\_OPENRAW is used. Other formats use the frequency determined by the file format.

*format*

[in] Optional. Specify 0 or FMOD\_SOUND\_FORMAT\_NONE to ignore. Format of the sound mandatory if FMOD\_OPENUSER or FMOD\_OPENRAW is used. Other formats use the format determined by the file format.

*decodebuffersize*

[in] Optional. Specify 0 to ignore. For streams. This determines the size of the double buffer (in PCM samples) that a stream uses. Use this for user created streams if you want to determine the size of the callback buffer passed to you. Specify 0 to use FMOD's default size which is currently equivalent to 400ms of the sound format created/loaded.

*initialsubsound*

[in] Optional. Specify 0 to ignore. In a multi-sample file format such as .FSB/.DLS/.SF2, specify the initial subsound to seek to, only if FMOD\_CREATESTREAM is used.

*numsubsounds*

[in] Optional. Specify 0 to ignore or have no subsounds. In a user created multi-sample sound, specify the number of subsounds within the sound that are accessible with Sound::getSubSound.

*inclusionlist*

[in] Optional. Specify 0 to ignore. In a multi-sample format such as .FSB/.DLS/.SF2 it may be desirable to specify only a subset of sounds to be loaded out of the whole file. This is an array of subsound indices to load into memory when created.

*inclusionlistnum*

[in] Optional. Specify 0 to ignore. This is the number of integers contained within the inclusionlist array.

#### *pcmreadcallback*

[in] Optional. Specify 0 to ignore. Callback to 'piggyback' on FMOD's read functions and accept or even write PCM data while FMOD is opening the sound. Used for user sounds created with FMOD\_OPENUSER or for capturing decoded data as FMOD reads it.

#### *pcmsetposcallback*

[in] Optional. Specify 0 to ignore. Callback for when the user calls a seeking function such as Channel::setTime or Channel::setPosition within a multi-sample sound, and for when it is opened.

#### *nonblockcallback*

[in] Optional. Specify 0 to ignore. Callback for successful completion, or error while loading a sound that used the FMOD\_NONBLOCKING flag.

#### *dlsname*

[in] Optional. Specify 0 to ignore. Filename for a DLS or SF2 sample set when loading a MIDI file. If not specified, on Windows it will attempt to open /windows/system32/drivers/gm.dls or /windows/system32/drivers/etc/gm.dls, on Mac it will attempt to load /System/Library/Components/CoreAudio.component/Contents/Resources/gs\_instruments.dls, otherwise the MIDI will fail to open. Current DLS support is for level 1 of the specification.

#### *encryptionkey*

[in] Optional. Specify 0 to ignore. Key for encrypted FSB file. Without this key an encrypted FSB file will not load.

#### *maxpolyphony*

[in] Optional. Specify 0 to ignore. For sequenced formats with dynamic channel allocation such as .MID and .IT, this specifies the maximum voice count allowed while playing. .IT defaults to 64. .MID defaults to 32.

#### *userdata*

[in] Optional. Specify 0 to ignore. This is user data to be attached to the sound during creation. Access via Sound::getUserData.

#### *suggestedsoundtype*

[in] Optional. Specify 0 or FMOD\_SOUND\_TYPE\_UNKNOWN to ignore. Instead of scanning all codec types, use this to speed up loading by making it jump straight to this codec.

#### *useropen*

[in] Optional. Specify 0 to ignore. Callback for opening this file.

#### *userclose*

[in] Optional. Specify 0 to ignore. Callback for closing this file.

#### *userread*

[in] Optional. Specify 0 to ignore. Callback for reading from this file.

*userseek*

[in] Optional. Specify 0 to ignore. Callback for seeking within this file.

*speakermap*

[in] Optional. Specify 0 to ignore. Use this to differ the way fmod maps multichannel sounds to speakers. See [FMOD\\_SPEAKERMATYPE](#) for more.

*initialsoundgroup*

[in] Optional. Specify 0 to ignore. Specify a sound group if required, to put sound in as it is created.

*initialseekposition*

[in] Optional. Specify 0 to ignore. For streams. Specify an initial position to seek the stream to.

*initialseekpostype*

[in] Optional. Specify 0 to ignore. For streams. Specify the time unit for the position set in initialseekposition.

## Remarks

This structure is optional! Specify 0 or NULL in [System::createSound](#) if you don't need it!

Members marked with [in] mean the user sets the value before passing it to the function.  
Members marked with [out] mean FMOD sets the value to be used after the function exits.

### Loading a file from memory.

- Create the sound using the FMOD\_OPENMEMORY flag.
- Mandatory. Specify 'length' for the size of the memory block in bytes.
- Other flags are optional.

### Loading a file from within another larger (possibly wad/pak) file, by giving the loader an offset and length.

- Mandatory. Specify 'fileoffset' and 'length'.
- Other flags are optional.

### To create a user created / non file based sound.

- Create the sound using the FMOD\_OPENUSER flag.
- Mandatory. Specify 'defaultfrequency', 'numchannels' and 'format'.
- Other flags are optional.

### To specify a starting subsound to seek to and flush with, within a multi-sample stream (ie FSB/DLS/SF2).

- Mandatory. Specify 'initialsubsound'.

### To specify which subsounds to load for multi-sample sounds (ie FSB/DLS/SF2) so that memory is saved and only a subset is actually loaded/read from disk.

- Mandatory. Specify 'inclusionlist' and 'inclusionlistnum'.

To specify 'piggyback' read and seek callbacks for capture of sound data as fmod reads and decodes it. Useful for ripping decoded PCM data from sounds as they are loaded / played.

- Mandatory. Specify 'pcmreadcallback' and 'pcmseekcallback'.

To specify a MIDI DLS/SF2 sample set file to load when opening a MIDI file.

- Mandatory. Specify 'dlsname'.

Setting the 'decodebuffersize' is for cpu intensive codecs that may be causing stuttering, not file intensive codecs (ie those from CD or netstreams) which are normally altered with [System::setStreamBufferSize](#). As an example of cpu intensive codecs, an mp3 file will take more cpu to decode than a PCM wav file.

If you have a stuttering effect, then it is using more cpu than the decode buffer playback rate can keep up with.

Increasing the decode buffersize will most likely solve this problem.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [System::createSound](#)
- [System::setStreamBufferSize](#)
- [FMOD\\_MODE](#)
- [FMOD\\_SOUND\\_FORMAT](#)
- [FMOD\\_SOUND\\_TYPE](#)
- [FMOD\\_SPEAKERMAPTYPE](#)

# FMOD\_DSP\_DESCRIPTION

When creating a DSP unit, declare one of these and provide the relevant callbacks and name for FMOD to use when it creates and uses a DSP unit of this type.?

## Structure

```

type fs tnc t{
char name[ 3 2;
unsigned int wsize n
int chane b;

MOD BSPC RA TCA LLACK c ea e;
MOD BSP E EASECA LLACK e base;
MOD BSP RSE TA LLACK ese t
MOD BSP RA DA LLACK ea d
MOD BSPSE TDSI TO NA LLACK se tpsi fo n

int nmp ame e s;
MOD BSP A AME T RESC * p am dsc;
MOD BSPSE TA AMCA LLACK se tp ame e r
MOD BSPGE TA AMCA LLACK ge tp ame e r
MOD BSP DA DGCA LLACK co nfg;

int co nfgwi dth
int co nfg big ht
wid * use rd a;
} MOD BSP HSC E PTO N

```

## Members

*name*

[in] Name of the unit to be displayed in the network.

*version*

[in] Plugin writer's version number.

*channels*

[in] Number of channels. Use 0 to process whatever number of channels is currently in the network. >0 would be mostly used if the unit is a unit that only generates sound.

*create*

[in] Create callback. This is called when DSP unit is created. Can be null.

*release*

[in] Release callback. This is called just before the unit is freed so the user can do any cleanup needed for the unit. Can be null.

*reset*

[in] Reset callback. This is called by the user to reset any history buffers that may need resetting for a filter, when it is to be used or re-used for the first time to its initial clean state. Use to avoid clicks or artifacts.

*read*

[in] Read callback. Processing is done here. Can be null.

*setposition*

[in] Set position callback. This is called if the unit wants to update its position info but not process data, or reset a cursor position internally if it is reading data from a certain source. Can be null.

*numparameters*

[in] Number of parameters used in this filter. The user finds this with DSP::getNumParameters

*paramdesc*

[in] Variable number of parameter structures.

*setParameter*

[in] This is called when the user calls DSP::setParameter. Can be null.

*getParameter*

[in] This is called when the user calls DSP::getParameter. Can be null.

*config*

[in] This is called when the user calls DSP::showConfigDialog. Can be used to display a dialog to configure the filter. Can be null.

*configwidth*

[in] Width of config dialog graphic if there is one. 0 otherwise.

*configheight*

[in] Height of config dialog graphic if there is one. 0 otherwise.

*userdata*

[in] Optional. Specify 0 to ignore. This is user data to be attached to the DSP unit during creation. Access via DSP::getUserData.

## Remarks

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

There are 2 different ways to change a parameter in this architecture.

One is to use DSP::setParameter / DSP::getParameter. This is platform independent and is dynamic, so new unknown



plugins can have their parameters enumerated and used.

The other is to use `DSP::showConfigDialog`. This is platform specific and requires a GUI, and will display a dialog box to configure the plugin.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## **See Also**

- [\\_System::createDSP](#)
- [FMOD\\_DSP\\_STATE](#)

# FMOD\_DSP\_PARAMETERDES

Structure to define a parameter for a DSP unit.?

## Structure

```
typedef struct {
    float min;
    float max;
    float defaultval;
    char name[16];
    char label[16];
    const char * description;
} FMOD_DSP_PARAMETERDES;
```

## Members

*min*

[in] Minimum value of the parameter (ie 100.0).

*max*

[in] Maximum value of the parameter (ie 22050.0).

*defaultval*

[in] Default value of parameter.

*name*

[in] Name of the parameter to be displayed (ie "Cutoff frequency").

*label*

[in] Short string to be put next to value to denote the unit type (ie "hz").

*description*

[in] Description of the parameter to be displayed as a help item / tooltip for this parameter.

## Remarks

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

The step parameter tells the gui or application that the parameter has a certain granularity.

For example in the example of cutoff frequency with a range from 100.0 to 22050.0 you might only want the selection to be in 10hz increments. For this you would simply use 10.0 as the step value.

For a boolean, you can use min = 0.0, max = 1.0, step = 1.0. This way the only possible values are 0.0 and 1.0.

Some applications may detect min = 0.0, max = 1.0, step = 1.0 and replace a graphical slider bar with a checkbox instead.

A step value of 1.0 would simulate integer values only.

A step value of 0.0 would mean the full floating point range is accessible.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## **See Also**

- [\\_System::createDSP](#)
- [\\_DSP::setParameter](#)

# FMOD\_DSP\_STATE

DSP plugin structure that is passed into each callback.?

## Structure

```
typedef struct {  
    FMOD_DSP * instance;  
    void * plugin_data;  
} FMOD_DSP_STATE;
```

## Members

*instance*

[out] Handle to the DSP handle the user created. Not to be modified. C++ users cast to FMOD::DSP to use.

*plugin\_data*

[in] Plugin writer created data the output author wants to attach to this object.

## Remarks

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_DSP\\_DESCRIPTION](#)

# FMOD\_GUID

Structure describing a globally unique identifier.?

## Structure

```
typedef struct {
    unsigned int  Data1;
    unsigned short Data2;
    unsigned short Data3;
    unsigned char  Data4[8];
} FMOD_GUID;
```

## Members

### *Data1*

Specifies the first 8 hexadecimal digits of the GUID

### *Data2*

Specifies the first group of 4 hexadecimal digits.

### *Data3*

Specifies the second group of 4 hexadecimal digits.

### *Data4*

Array of 8 bytes. The first 2 bytes contain the third group of 4 hexadecimal digits. The remaining 6 bytes contain the final 12 hexadecimal digits.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getDriverInfo](#)

# FMOD\_OUTPUT\_DESCRIPTION

When creating an output, declare one of these and provide the relevant callbacks and name for FMOD to use when it opens and reads a file of this type.?

## Structure

```
typedef struct {
    const char * name;
    unsigned int version;
    int polling;
    FMOD_OUTPUTGETNUMDRIVERSCALLBACK getnumdrivers;
    FMOD_OUTPUTGETDRIVERNAMECALLBACK getdrivername;
    FMOD_OUTPUTGETDRIVERCAPSCALLBACK getdrivercaps;
    FMOD_OUTPUTTINTECALLBACK tint;
    FMOD_OUTPUTCLOCKCALLBACK clock;
    FMOD_OUTPUTTUPPERCASECALLBACK uppercase;
    FMOD_OUTPUTGETTHROTTLECALLBACK getthrottle;
    FMOD_OUTPUTGETPSIPOSITIONCALLBACK getpsi position;
    FMOD_OUTPUTTLOCKCALLBACK lock;
    FMOD_OUTPUTTUNLOCKCALLBACK unlock;
} FMOD_OUTPUT_DESCRIPTION
```

## Members

*name*

[in] Name of the output.

*version*

[in] Plugin writer's version number.

*polling*

[in] If TRUE (non zero), this tells FMOD to start a thread and call getposition / lock / unlock for feeding data. If 0, the output is probably callback based, so all the plugin needs to do is call readfrommixer to the appropriate pointer.

*getnumdrivers*

[in] For sound device enumeration. This callback is to give System::getNumDrivers something to return.

*getdrivername*

[in] For sound device enumeration. This callback is to give System::getDriverName something to return.

*getdrivercaps*

[in] For sound device enumeration. This callback is to give System::getDriverCaps something to return.

*init*

[in] Initialization function for the output device. This is called from System::init.

*close*

[in] Cleanup / close down function for the output device. This is called from System::close.

*update*

[in] Update function that is called once a frame by the user. This is called from System::update.

*gethandle*

[in] This is called from System::getOutputHandle. This is just to return a pointer to the internal system device object that the system may be using.

*getposition*

[in] This is called from the FMOD software mixer thread if 'polling' = true. This returns a position value in samples so that FMOD knows where and when to fill its buffer.

*lock*

[in] This is called from the FMOD software mixer thread if 'polling' = true. This function provides a pointer to data that FMOD can write to when software mixing.

*unlock*

[in] This is called from the FMOD software mixer thread if 'polling' = true. This optional function accepts the data that has been mixed and copies it or does whatever it needs to before sending it to the hardware.

## Remarks

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_OUTPUT\\_STATE](#)





# FMOD\_OUTPUT\_STATE

Output plugin structure that is passed into each callback.?

## Structure

```
typedef struct {  
    void * plugin_data;  
    FMOD_OUTPUT_STATE_READFROMMIXER readfrommixer;  
} FMOD_OUTPUT_STATE;
```

## Members

*plugin\_data*

[in] Plugin writer created data the output author wants to attach to this object.

*readfrommixer*

[out] Function to update mixer and write the result to the provided pointer. Used from callback based output only. Polling based output uses lock/unlock/getposition.

## Remarks

Members marked with [in] mean the variable can be written to. The user can set the value.

Members marked with [out] mean the variable is modified by FMOD and is for reading purposes only. Do not change this value.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_OUTPUT\\_DESCRIPTION](#)

# FMOD\_REVERB\_CHANNELPROPERTIES

Structure defining the properties for a reverb source, related to a FMOD channel.

?

For more indepth descriptions of the reverb properties under win32, please see the EAX3 documentation at <http://developer.creative.com/> under the 'downloads' section.

If they do not have the EAX3 documentation, then most information can be attained from the EAX2 documentation, as EAX3 only adds some more parameters and functionality on top of EAX2.

?

Note the default reverb properties are the same as the FMOD\_PRESET\_GENERIC preset.

Note that integer values that typically range from -10,000 to 1000 are represented in decibels, and are of a logarithmic scale, not linear, whereas float values are typically linear.

PORTABILITY: Each member has the platform it supports in braces ie (win32/Xbox).

Some reverb parameters are only supported in win32 and some only on Xbox. If all parameters are set then the reverb should product a similar effect on either platform.

?

The numerical values listed below are the maximum, minimum and default values for each variable respectively.

?

## Structure

```
typedef struct {
    int Direct;
    int DirectHF;
    int Room;
    int RoomHF;
    int Obstacle;
    float ObstacleLFRatio;
    int Occlusion;
    float OcclusionLFRatio;
    float OcclusionRoomRatio;
    float OcclusionDirectRatio;
    int Exclusion;
    float ExclusionLFRatio;
    int OutsideVolumeHF;
    float DampenReverb;
    float ReflectionReverb;
    float RoomReflectionReverb;
    float AirAbsorptionReverb;
    unsigned int Flags;
} FMOD_REVERB_CHANNELPROPERTIES;
```

## Members

### Direct

[in/out] -10000, 1000, 0, direct path level (at low and mid frequencies) (SUPPORTED:EAX/I3DL2/Xbox1/SFX)

### DirectHF

[in/out] -10000, 0, 0, relative direct path level at high frequencies (SUPPORTED:EAX/I3DL2/Xbox1)

### *Room*

[in/out] -10000, 1000, 0, room effect level (at low and mid frequencies)  
(SUPPORTED:EAX/I3DL2/Xbox1/GC/SFX)

### *RoomHF*

[in/out] -10000, 0, 0, relative room effect level at high frequencies (SUPPORTED:EAX/I3DL2/Xbox1)

### *Obstruction*

[in/out] -10000, 0, 0, main obstruction control (attenuation at high frequencies)  
(SUPPORTED:EAX/I3DL2/Xbox1)

### *ObstructionLFRatio*

[in/out] 0.0, 1.0, 0.0, obstruction low-frequency level re. main control (SUPPORTED:EAX/I3DL2/Xbox1)

### *Occlusion*

[in/out] -10000, 0, 0, main occlusion control (attenuation at high frequencies) (SUPPORTED:EAX/I3DL2/Xbox1)

### *OcclusionLFRatio*

[in/out] 0.0, 1.0, 0.25, occlusion low-frequency level re. main control (SUPPORTED:EAX/I3DL2/Xbox1)

### *OcclusionRoomRatio*

[in/out] 0.0, 10.0, 1.5, relative occlusion control for room effect (SUPPORTED:EAX)

### *OcclusionDirectRatio*

[in/out] 0.0, 10.0, 1.0, relative occlusion control for direct path (SUPPORTED:EAX)

### *Exclusion*

[in/out] -10000, 0, 0, main exclusion control (attenuation at high frequencies) (SUPPORTED:EAX)

### *ExclusionLFRatio*

[in/out] 0.0, 1.0, 1.0, exclusion low-frequency level re. main control (SUPPORTED:EAX)

### *OutsideVolumeHF*

[in/out] -10000, 0, 0, outside sound cone level at high frequencies (SUPPORTED:EAX)

### *DopplerFactor*

[in/out] 0.0, 10.0, 0.0, like DS3D flDopplerFactor but per source (SUPPORTED:EAX)

### *RolloffFactor*

[in/out] 0.0, 10.0, 0.0, like DS3D flRolloffFactor but per source (SUPPORTED:EAX)

### *RoomRolloffFactor*

[in/out] 0.0, 10.0, 0.0, like DS3D flRolloffFactor but for room effect (SUPPORTED:EAX/I3DL2/Xbox1)

### *AirAbsorptionFactor*

[in/out] 0.0, 10.0, 1.0, multiplies AirAbsorptionHF member of FMOD\_REVERB\_PROPERTIES (SUPPORTED:EAX)

### *Flags*

[in/out] [FMOD\\_REVERB\\_CHANNELFLAGS](#) - modifies the behavior of properties (SUPPORTED:EAX)

## Remarks

**SUPPORTED** next to each parameter means the platform the parameter can be set on. Some platforms support all parameters and some don't.

EAX means hardware reverb on FMOD\_OUTPUTTYPE\_DSOUND on windows only (must use FMOD\_HARDWARE), on soundcards that support EAX 1 to 4.

EAX4 means hardware reverb on FMOD\_OUTPUTTYPE\_DSOUND on windows only (must use FMOD\_HARDWARE), on soundcards that support EAX 4.

I3DL2 means hardware reverb on FMOD\_OUTPUTTYPE\_DSOUND on windows only (must use FMOD\_HARDWARE), on soundcards that support I3DL2 non EAX native reverb.

GC means Nintendo Gamecube hardware reverb (must use FMOD\_HARDWARE).

WII means Nintendo Wii hardware reverb (must use FMOD\_HARDWARE).

Xbox1 means the original Xbox hardware reverb (must use FMOD\_HARDWARE).

PS2 means Playstation 2 hardware reverb (must use FMOD\_HARDWARE).

SFX means FMOD SFX software reverb. This works on any platform that uses FMOD\_SOFTWARE for loading sounds.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [Channel::setReverbProperties](#)
- [Channel::getReverbProperties](#)
- [FMOD\\_REVERB\\_CHANNELFLAGS](#)

# FMOD\_REVERB\_PROPERTIES

Structure defining a reverb environment.

?

?For more indepth descriptions of the reverb properties under win32, please see the EAX2 and EAX3?documentation at <http://developer.creative.com/> under the 'downloads' section.

?If they do not have the EAX3 documentation, then most information can be attained from?the EAX2 documentation, as EAX3 only adds some more parameters and functionality on top of?EAX2.?

## Structure

```

y p d f s t n c t {
i n t I s a n e;
i n t E n v o m e n t
    f b a t E n s i e;
    f b a t E n v D f f i s i o n
i n t B o m;
i n t B o m H F
i n t B o m L F
    f b a t B c a y T m e;
    f b a t B c a y H F a t o;
    f b a t B c a y L F a t o;
i n t A f e c t o s;
    f b a t A f e c t o s B h y;
    f b a t A f e c t o s A h 3];
i n t A e r b
    f b a t A e r b B h y;
    f b a t A e r b A h 3];
    f b a t E c h T m e;
    f b a t E c h B p t h
    f b a t M o d h t o n T m e;
    f b a t M o d h t o n B p t h
    f b a t A i A b o r p t o n H F
    f b a t H F a e e n e;
    f b a t L F a e e n e;
    f b a t B o m A l b f f e c t o r
    f b a t D f f i s i o n
    f b a t B s i y;
u s i g a d i n t F l a g s;
} M O D E X R B P R E R T E S;

```

## Members

*Instance*

[in] 0, 3, 0, Environment Instance. Simultaneous HW reverbs are possible on some platforms. (SUPPORTED:EAX4(3 instances)/GC and Wii (2 instances))

## Environment

[in/out] -1 , 25 , -1 , sets all listener properties. -1 = OFF. (SUPPORTED:EAX/PS2)

$$EnvSize$$

[in/out] 1.0 , 100.0 , 7.5 , environment size in meters (SUPPORTED:EAX)

### *EnvDiffusion*

[in/out] 0.0 , 1.0 , 1.0 , environment diffusion (SUPPORTED:EAX/Xbox1/GC)

### *Room*

[in/out] -10000, 0 , -1000 , room effect level (at mid frequencies) (SUPPORTED:EAX/Xbox1/GC/I3DL2/SFX)

### *RoomHF*

[in/out] -10000, 0 , -100 , relative room effect level at high frequencies (SUPPORTED:EAX/Xbox1/I3DL2/SFX)

### *RoomLF*

[in/out] -10000, 0 , 0 , relative room effect level at low frequencies (SUPPORTED:EAX/SFX)

### *DecayTime*

[in/out] 0.1 , 20.0 , 1.49 , reverberation decay time at mid frequencies  
(SUPPORTED:EAX/Xbox1/GC/I3DL2/SFX)

### *DecayHFRatio*

[in/out] 0.1 , 2.0 , 0.83 , high-frequency to mid-frequency decay time ratio  
(SUPPORTED:EAX/Xbox1/I3DL2/SFX)

### *DecayLFRatio*

[in/out] 0.1 , 2.0 , 1.0 , low-frequency to mid-frequency decay time ratio (SUPPORTED:EAX)

### *Reflections*

[in/out] -10000, 1000 , -2602 , early reflections level relative to room effect  
(SUPPORTED:EAX/Xbox1/GC/I3DL2/SFX)

### *ReflectionsDelay*

[in/out] 0.0 , 0.3 , 0.007 , initial reflection delay time (SUPPORTED:EAX/Xbox1/I3DL2/SFX)

### *ReflectionsPan*

[in/out] , , [0,0,0], early reflections panning vector (SUPPORTED:EAX)

### *Reverb*

[in/out] -10000, 2000 , 200 , late reverberation level relative to room effect  
(SUPPORTED:EAX/Xbox1/I3DL2/SFX)

### *ReverbDelay*

[in/out] 0.0 , 0.1 , 0.011 , late reverberation delay time relative to initial reflection  
(SUPPORTED:EAX/Xbox1/GC/I3DL2/SFX)

### *ReverbPan*

[in/out] , , [0,0,0], late reverberation panning vector (SUPPORTED:EAX)

### *EchoTime*

[in/out] .075 , 0.25 , 0.25 , echo time  
(SUPPORTED:EAX/PS2(FMOD\_PRESET\_PS2\_ECHO/FMOD\_PRESET\_PS2\_DELAY only)

### *EchoDepth*

[in/out] 0.0 , 1.0 , 0.0 , echo depth (SUPPORTED:EAX/PS2(FMOD\_PRESET\_PS2\_ECHO only)

### *ModulationTime*

[in/out] 0.04 , 4.0 , 0.25 , modulation time (SUPPORTED:EAX)

### *ModulationDepth*

[in/out] 0.0 , 1.0 , 0.0 , modulation depth (SUPPORTED:EAX/GC)

### *AirAbsorptionHF*

[in/out] -100 , 0.0 , -5.0 , change in level per meter at high frequencies (SUPPORTED:EAX)

### *HFRference*

[in/out] 1000.0, 20000 , 5000.0 , reference high frequency (hz) (SUPPORTED:EAX/Xbox1/I3DL2/SFX)

### *LFReference*

[in/out] 20.0 , 1000.0, 250.0 , reference low frequency (hz) (SUPPORTED:EAX/SFX)

### *RoomRolloffFactor*

[in/out] 0.0 , 10.0 , 0.0 , like rolloffscale in System::set3DSettings but for reverb room size effect  
(SUPPORTED:EAX/Xbox1/I3DL2/SFX)

### *Diffusion*

[in/out] 0.0 , 100.0 , 100.0 , Value that controls the echo density in the late reverberation decay.  
(SUPPORTED:I3DL2/Xbox1/SFX)

### *Density*

[in/out] 0.0 , 100.0 , 100.0 , Value that controls the modal density in the late reverberation decay  
(SUPPORTED:I3DL2/Xbox1/SFX)

### *Flags*

[in/out] [FMOD\\_REVERB\\_FLAGS](#) - modifies the behavior of above properties  
(SUPPORTED:EAX/PS2/GC/WII)

### **Remarks**

Note the default reverb properties are the same as the FMOD\_PRESET\_GENERIC preset.

Note that integer values that typically range from -10,000 to 1000 are represented in decibels, and are of a logarithmic scale, not linear, whereas float values are always linear.

The numerical values listed below are the maximum, minimum and default values for each variable respectively.

**SUPPORTED** next to each parameter means the platform the parameter can be set on. Some platforms support all parameters and some don't.

EAX means hardware reverb on FMOD\_OUTPUTTYPE\_DSOUND on windows only (must use FMOD\_HARDWARE), on soundcards that support EAX 1 to 4.

EAX4 means hardware reverb on FMOD\_OUTPUTTYPE\_DSOUND on windows only (must use FMOD\_HARDWARE), on soundcards that support EAX 4.

I3DL2 means hardware reverb on FMOD\_OUTPUTTYPE\_DSOUND on windows only (must use FMOD\_HARDWARE), on soundcards that support I3DL2 non EAX native reverb.

GC means Nintendo Gamecube hardware reverb (must use FMOD\_HARDWARE).

WII means Nintendo Wii hardware reverb (must use FMOD\_HARDWARE).

Xbox1 means the original Xbox hardware reverb (must use FMOD\_HARDWARE).

PS2 means Playstation 2 hardware reverb (must use FMOD\_HARDWARE).

SFX means FMOD SFX software reverb. This works on any platform that uses FMOD\_SOFTWARE for loading sounds.

Members marked with [in] mean the user sets the value before passing it to the function.

Members marked with [out] mean FMOD sets the value to be used after the function exits.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setReverbProperties](#)
- [System::getReverbProperties](#)
- [FMOD\\_REVERB\\_PRESETS](#)
- [FMOD\\_REVERB\\_FLAGS](#)



# FMOD\_TAG

Structure describing a piece of tag data.?

## Structure

```
typedef struct {
    FMOD_TAG_TYPE type;
    FMOD_TAG_DATA_TYPE data_type;
    char * name;
    void * data;
    unsigned int datalen;
    FMOD_BOOL updated;
} FMOD_TAG;
```

## Members

*type*

[out] The type of this tag.

*datatype*

[out] The type of data that this tag contains

*name*

[out] The name of this tag i.e. "TITLE", "ARTIST" etc.

*data*

[out] Pointer to the tag data - its format is determined by the datatype member

*datalen*

[out] Length of the data contained in this tag

*updated*

[out] True if this tag has been updated since last being accessed with [Sound::getTag](#)

## Remarks

Members marked with [in] mean the user sets the value before passing it to the function.  
Members marked with [out] mean FMOD sets the value to be used after the function exits.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

### See Also

- [Sound::getTag](#)
- [FMOD\\_TAGTYPE](#)
- [FMOD\\_TAGDATATYPE](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_VECTOR

Structure describing a point in 3D space.?

## Structure

```
typedef struct {  
    float x;  
    float y;  
    float z;  
} FMOD_VECTOR;
```

## Members

*x*

X co-ordinate in 3D space.

*y*

Y co-ordinate in 3D space.

*z*

Z co-ordinate in 3D space.

## Remarks

FMOD uses a left handed co-ordinate system by default.

To use a right handed co-ordinate system specify FMOD\_INIT\_3D\_RIGHTHANDED from [FMOD\\_INITFLAGS](#) in System::init.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::set3DListenerAttributes](#)
- [System::get3DListenerAttributes](#)
- [Channel::set3DAttributes](#)
- [Channel::get3DAttributes](#)
- [Channel::set3DCustomRolloff](#)
- [Channel::get3DCustomRolloff](#)
- [Sound::set3DCustomRolloff](#)

- [\\_Sound::get3DCustomRolloff](#)
- [\\_Geometry::addPolygon](#)
- [\\_Geometry::setPolygonVertex](#)
- [\\_Geometry::getPolygonVertex](#)
- [\\_Geometry::setRotation](#)
- [\\_Geometry::getRotation](#)
- [\\_Geometry::setPosition](#)
- [\\_Geometry::getPosition](#)
- [\\_Geometry::setScale](#)
- [\\_Geometry::getScale](#)
- [\\_EMOD\\_INITFLAGS](#)

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# Defines

[FMOD\\_CAPS](#)

[FMOD\\_DEBUGLEVEL](#)

[FMOD\\_INITFLAGS](#)

[FMOD\\_MEMORY\\_TYPE](#)

[FMOD\\_MODE](#)

[FMOD\\_REVERB\\_CHANNELFLAGS](#)

[FMOD\\_REVERB\\_FLAGS](#)

[FMOD\\_REVERB\\_PRESETS](#)

[FMOD\\_TIMEUNIT](#)

# FMOD\_CAPS

Bit fields to use with [System::getDriverCaps](#) to determine the capabilities of a card / output device. It is important to check FMOD\_CAPS\_HARDWARE\_EMULATED on windows machines, to then adjust [System::setDSPBufferSize](#) to (1024, 10) to compensate for the higher latency.

## Definition

```
#define FMOD_CAPS_NONE 0x00000000
#define FMOD_CAPS_HARDWARE 0x00000001
#define FMOD_CAPS_HARDWARE_EMULATED 0x00000002
#define FMOD_CAPS_OUTPUT_MULTICHANNEL 0x00000004
#define FMOD_CAPS_OUTPUT_FORMAT_PCM8 0x00000008
#define FMOD_CAPS_OUTPUT_FORMAT_PCM16 0x00000010
#define FMOD_CAPS_OUTPUT_FORMAT_PCM24 0x00000020
#define FMOD_CAPS_OUTPUT_FORMAT_PCM32 0x00000040
#define FMOD_CAPS_OUTPUT_FORMAT_FLOAT 0x00000080
#define FMOD_CAPS_EXAMPLE_RATE_2 0x00000100
#define FMOD_CAPS_EXAMPLE_RATE_3 0x00000200
#define FMOD_CAPS_EXAMPLE_RATE_4 0x00000400
#define FMOD_CAPS_EXAMPLE_RATE_8 0x00000800
#define FMOD_CAPS_EXAMPLE_RATE_13 0x00001000
#define FMOD_CAPS_EXAMPLE_RATE_16 0x00002000
```

## Values

*FMOD\_CAPS\_NONE*

Device has no special capabilities.

*FMOD\_CAPS\_HARDWARE*

Device supports hardware mixing.

*FMOD\_CAPS\_HARDWARE\_EMULATED*

User has device set to 'Hardware acceleration = off' in control panel, and now extra 200ms latency is incurred.

*FMOD\_CAPS\_OUTPUT\_MULTICHANNEL*

Device can do multichannel output, ie greater than 2 channels.

*FMOD\_CAPS\_OUTPUT\_FORMAT\_PCM8*

Device can output to 8bit integer PCM.

*FMOD\_CAPS\_OUTPUT\_FORMAT\_PCM16*

Device can output to 16bit integer PCM.

*FMOD\_CAPS\_OUTPUT\_FORMAT\_PCM24*

Device can output to 24bit integer PCM.

*FMOD\_CAPS\_OUTPUT\_FORMAT\_PCM32*

Device can output to 32bit integer PCM.

*FMOD\_CAPS\_OUTPUT\_FORMAT\_PCMFLOAT*

Device can output to 32bit floating point PCM.

*FMOD\_CAPS\_REVERB\_EAX2*

Device supports EAX2 reverb.

*FMOD\_CAPS\_REVERB\_EAX3*

Device supports EAX3 reverb.

*FMOD\_CAPS\_REVERB\_EAX4*

Device supports EAX4 reverb

*FMOD\_CAPS\_REVERB\_EAX5*

Device supports EAX5 reverb

*FMOD\_CAPS\_REVERB\_I3DL2*

Device supports I3DL2 reverb.

*FMOD\_CAPS\_REVERB\_LIMITED*

Device supports some form of limited hardware reverb, maybe parameterless and only selectable by environment.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_System::getDriverCaps](#)
- [\\_System::setDSPBufferSize](#)

# FMOD\_DEBUGLEVEL

Bit fields to use with FMOD::Debug\_SetLevel / FMOD::Debug\_GetLevel to control the level of tty debug output with logging versions of FMOD (fmodL).?

## Definition

```
# d f a F M O D _ E R G _ E V L N E 0x00000000
# d f a F M O D _ E R G _ E V L D G 0x00000001
# d f a F M O D _ E R G _ E V L E R R 0x00000002
# d f a F M O D _ E R G _ E V L W A R N I N G 0x00000004
# d f a F M O D _ E R G _ E V L H I N T 0x00000008
# d f a F M O D _ E R G _ E V L A L L 0x000000FF
# d f a F M O D _ E R G _ T E M E M O R Y 0x00000100
# d f a F M O D _ E R G _ T E T H R E A D 0x00000200
# d f a F M O D _ E R G _ T E F I L E 0x00000400
# d f a F M O D _ E R G _ T E N E T 0x00000800
# d f a F M O D _ E R G _ T E E V E N T 0x000001000
# d f a F M O D _ E R G _ T E A L L 0x0000FFFF
# d f a F M O D _ E R G _ M S P A Y _ T M E S T A M P 0x01000000
# d f a F M O D _ E R G _ M S P A Y _ L I N E N U M B E R 0x02000000
# d f a F M O D _ E R G _ M S P A Y _ C O M P R E S S 0x04000000
# d f a F M O D _ E R G _ M S P A Y _ A L L 0x0F000000
# d f a F M O D _ E R G _ A L L 0xFFFFFFFF
```

## Values

*FMOD\_DEBUG\_LEVEL\_NONE*

*FMOD\_DEBUG\_LEVEL\_LOG*

*FMOD\_DEBUG\_LEVEL\_ERROR*

*FMOD\_DEBUG\_LEVEL\_WARNING*

*FMOD\_DEBUG\_LEVEL\_HINT*

*FMOD\_DEBUG\_LEVEL\_ALL*

*FMOD\_DEBUG\_TYPE\_MEMORY*

*FMOD\_DEBUG\_TYPE\_THREAD*

*FMOD\_DEBUG\_TYPE\_FILE*



*FMOD\_DEBUG\_TYPE\_NET*

*FMOD\_DEBUG\_TYPE\_EVENT*

*FMOD\_DEBUG\_TYPE\_ALL*

*FMOD\_DEBUG\_DISPLAY\_TIMESTAMPS*

*FMOD\_DEBUG\_DISPLAY\_LINENUMBERS*

*FMOD\_DEBUG\_DISPLAY\_COMPRESS*

*FMOD\_DEBUG\_DISPLAY\_ALL*

*FMOD\_DEBUG\_ALL*

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [Debug\\_SetLevel](#)
- [Debug\\_GetLevel](#)

# FMOD\_INITFLAGS

Initialization flags. Use them with [System::init](#) in the flags parameter to change various behavior. Use [System::setAdvancedSettings](#) to adjust settings for some of the features that are enabled by these flags.

## Definition

```
#define FMOD_INIT_NORMAL 0x00000000
#define FMOD_INIT_STREAM_FROM_UPDATE 0x00000001
#define FMOD_INIT_3D_RIGHTHANDED 0x00000002
#define FMOD_INIT_SOFTWARE_DISABLE 0x00000004
#define FMOD_INIT_SOFTWARE_OCCLUSION 0x00000008
#define FMOD_INIT_DSP_100 0x00000010
#define FMOD_INIT_DSP_20 0x00000020
#define FMOD_INIT_DSP_50 0x00000040
#define FMOD_INIT_DSP_10 0x00000080
#define FMOD_INIT_DSP_5 0x00000100
#define FMOD_INIT_DSP_2 0x00000200
#define FMOD_INIT_DSP_1 0x00000400
#define FMOD_INIT_DSP_0 0x00000800
#define FMOD_INIT_B2_DISABLE 0x00010000
#define FMOD_INIT_B2_DISABLE1 0x00020000
#define FMOD_INIT_B2_DISABLE2 0x00040000
#define FMOD_INIT_B2_DISABLE3 0x00080000
#define FMOD_INIT_B2_DISABLE4 0x00100000
#define FMOD_INIT_B2_DISABLE5 0x00200000
#define FMOD_INIT_B2_DISABLE6 0x00400000
#define FMOD_INIT_B2_DISABLE7 0x00800000
#define FMOD_INIT_B2_DISABLE8 0x01000000
#define FMOD_INIT_B2_DISABLE9 0x02000000
#define FMOD_INIT_B2_DISABLE10 0x04000000
#define FMOD_INIT_B2_DISABLE11 0x08000000
#define FMOD_INIT_B2_DISABLE12 0x10000000
#define FMOD_INIT_B2_DISABLE13 0x20000000
#define FMOD_INIT_B2_DISABLE14 0x40000000
#define FMOD_INIT_B2_DISABLE15 0x80000000
```

## Values

*FMOD\_INIT\_NORMAL*

All platforms - Initialize normally

*FMOD\_INIT\_STREAM\_FROM\_UPDATE*

All platforms - No stream thread is created internally. Streams are driven from [System::update](#). Mainly used with non-realtime outputs.

*FMOD\_INIT\_3D\_RIGHTHANDED*

All platforms - FMOD will treat +X as right, +Y as up and +Z as backwards (towards you).

*FMOD\_INIT\_SOFTWARE\_DISABLE*

All platforms - Disable software mixer to save memory. Anything created with FMOD\_SOFTWARE will fail and DSP will not work.

*FMOD\_INIT\_SOFTWARE\_OCCLUSION*

All platforms - All FMOD\_SOFTWARE with FMOD\_3D based voices will add a software lowpass filter effect into the DSP chain which is automatically used when [Channel::set3DOcclusion](#) is used or the geometry API.

### *FMOD\_INIT\_SOFTWARE\_HRTF*

All platforms - All FMOD\_SOFTWARE with FMOD\_3D based voices will add a software lowpass filter effect into the DSP chain which causes sounds to sound duller when the sound goes behind the listener. Use [System::setAdvancedSettings](#) to adjust cutoff frequency.

### *FMOD\_INIT\_ENABLE\_DSPNET*

All platforms - Enable TCP/IP based host which allows "DSPNet Listener.exe" to connect to it, and view the DSP dataflow network graph in real-time.

### *FMOD\_INIT\_VOL0\_BECOMES\_VIRTUAL*

All platforms - Any sounds that are 0 volume will go virtual and not be processed except for having their positions updated virtually. Use [System::setAdvancedSettings](#) to adjust what volume besides zero to switch to virtual at.

### *FMOD\_INIT\_WASAPI\_EXCLUSIVE*

Win32 Vista only - for WASAPI output - Enable exclusive access to hardware, lower latency at the expense of excluding other applications from accessing the audio hardware.

### *FMOD\_INIT\_DSOUND\_HRTFNONE*

Win32 only - for DirectSound output - FMOD\_HARDWARE | FMOD\_3D buffers use simple stereo panning/doppler/attenuation when 3D hardware acceleration is not present.

### *FMOD\_INIT\_DSOUND\_HRTFLIGHT*

Win32 only - for DirectSound output - FMOD\_HARDWARE | FMOD\_3D buffers use a slightly higher quality algorithm when 3D hardware acceleration is not present.

### *FMOD\_INIT\_DSOUND\_HRTFFULL*

Win32 only - for DirectSound output - FMOD\_HARDWARE | FMOD\_3D buffers use full quality 3D playback when 3d hardware acceleration is not present.

### *FMOD\_INIT\_PS2\_DISABLECORE0REVERB*

PS2 only - Disable reverb on CORE 0 to regain 256k SRAM.

### *FMOD\_INIT\_PS2\_DISABLECORE1REVERB*

PS2 only - Disable reverb on CORE 1 to regain 256k SRAM.

### *FMOD\_INIT\_PS2\_DONTUSESCRATCHPAD*

PS2 only - Disable FMOD's usage of the scratchpad.

### *FMOD\_INIT\_PS2\_SWAPDMACHANNELS*

PS2 only - Changes FMOD from using SPU DMA channel 0 for software mixing, and 1 for sound data upload/file streaming, to 1 and 0 respectively.

### *FMOD\_INIT\_XBOX\_REMOVEHEADROOM*

Xbox only - By default DirectSound attenuates all sound by 6db to avoid clipping/distortion. CAUTION. If you use this flag you are responsible for the final mix to make sure clipping / distortion doesn't happen.

*FMOD\_INIT\_360\_MUSICMUTENOTPAUSE*

Xbox 360 only - The "music" channelgroup which by default pauses when custom 360 dashboard music is played, can be changed to mute (therefore continues playing) instead of pausing, by using this flag.

*FMOD\_INIT\_SYNCMIXERWITHUPDATE*

Win32/Wii/PS3/Xbox/Xbox 360 - FMOD Mixer thread is woken up to do a mix when [System:update](#) is called rather than waking periodically on its own timer.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_System:init](#)
- [\\_System:update](#)
- [\\_System:setAdvancedSettings](#)
- [\\_Channel:set3DOcclusion](#)

# FMOD\_MEMORY\_TYPE

Bit fields for memory allocation type being passed into FMOD memory callbacks.?

## Definition

```
# d e f n e F M O D _ M E M O R Y _ N O R M A L 0x00000000
# d e f n e F M O D _ M E M O R Y _ X B O X 3 6 0 _ P H Y S I C A L 0x00100000
# d e f n e F M O D _ M E M O R Y _ P E R S I S T E N T 0x00200000
```

## Values

*FMOD\_MEMORY\_NORMAL*

Standard memory.

*FMOD\_MEMORY\_XBOX360\_PHYSICAL*

Requires XPhysicalAlloc / XPhysicalFree.

*FMOD\_MEMORY\_PERSISTENT*

Persistent memory. Memory will be freed when System::release is called.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_MEMORY\\_ALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_REALLOCCALLBACK](#)
- [FMOD\\_MEMORY\\_FREECALLBACK](#)
- [Memory\\_Initialize](#)

# FMOD\_MODE

Sound description bitfields, bitwise OR them together for loading and describing sounds. By default a sound will open as a static sound that is decompressed fully into memory to PCM. (ie equivalent of FMOD\_CREATEAMPLE)

To have a sound stream instead, use FMOD\_CREATESTREAM, or use the wrapper function

[System::createStream](#).

Some opening modes (ie FMOD\_OPENUSER, FMOD\_OPENMEMORY, FMOD\_OPENMEMORY\_POINT, FMOD\_OPENRAW) will need extra information.

This can be provided using the FMOD\_CREATESOUNDEXINFO structure.

On Playstation 2, non VAG formats will default to FMOD\_SOFTWARE if FMOD\_HARDWARE is not specified.

This is due to PS2 hardware not supporting PCM data.

?

Specifying FMOD\_OPENMEMORY\_POINT will POINT to your memory rather allocating its own sound buffers and duplicating it internally.

**This means you cannot free the memory while FMOD is using it, until after Sound::release is called.** With

FMOD\_OPENMEMORY\_POINT, for PCM formats, only WAV, FSB, and RAW are supported. For compressed formats, only those formats supported by FMOD\_CREATECOMPRESSEDAMPLE are supported.

With FMOD\_OPENMEMORY\_POINT and FMOD\_OPENRAW, if using them together, note that you must pad the data on each side by 16 bytes. This is so fmod can modify the ends of the data for looping/interpolation/mixing purposes.

?

**Xbox 360 memory** On Xbox 360 Specifying FMOD\_OPENMEMORY\_POINT to a virtual memory address will cause FMOD\_ERR\_INVALID\_ADDRESS to be returned. Use physical memory only for this functionality.

?

FMOD\_LOWMEM is used on a sound if you want to minimize the memory overhead, by having FMOD not allocate memory for certain features that are not likely to be used in a game environment. These are :

1. Sound::getName functionality is removed. 256 bytes per sound is saved.

2. For a stream, a default sentence is not created, 4 bytes per subsound. On a 2000 subsound FSB this can save 8kb for example. Sound::setSubSoundSentence can simply be used to set up a sentence as normal, System::playSound just wont play through the whole set of subsounds by default any more.

?

## Definition

```
#define FMOD_DEFAULT 0x00000000
#define FMOD_DOPPOFF 0x00000001
#define FMOD_DOPNMA L 0x00000002
#define FMOD_DOPBID 0x00000004
#define FMOD_2D 0x00000008
#define FMOD_3D 0x00000010
#define FMOD_HARDWARE 0x00000020
#define FMOD_SOFTWARE 0x00000040
#define FMOD_CKATES TEAM 0x00000080
#define FMOD_CKATESAM PE 0x00000100
#define FMOD_CKATECOM PESSE SAM PE 0x00000200
#define FMOD_DOEUSE R 0x00000400
#define FMOD_DOE NEMO K 0x00000800
#define FMOD_DOE NEMO K_ DI NT 0x00001000
#define FMOD_DOE NAW 0x00002000
#define FMOD_DOE NNY 0x00004000
#define FMOD_DACCURATE ME 0x00008000
#define FMOD_DM EGSEA R H 0x00010000
```

```

# d f a FMOD_N NBDCKI N 0x0001 0000
# d f a FMOD_UNQUE 0x00020000
# d f a FMOD_3D_EA DE A T E 0x00040000
# d f a FMOD_3D_WO RLDE A T E 0x00080000
# d f a FMOD_3D_DG B LD FF 0x001 00000
# d f a FMOD_3D_I NA RB LD FF 0x00200000
# d f a FMOD_3D_DCUS VM B LD FF 0x04000000
# d f a FMOD_3D_DIG N EGEOME TE 0x40000000
# d f a FMOD_DC DA_ B REAS E 0x00400000
# d f a FMOD_DC DA_ J TE RO REC T 0x00800000
# d f a FMOD_UNCO E 0x1 000000
# d f a FMOD_DIG N E AGS 0x02000000
# d f a FMOD_DWMEM 0x08000000
# d f a FMOD_DA SECO NA E AM 0x20000000
# d f a FMOD_V RTA L PAY FRMS A RT 0x80000000

```

## Values

### *FMOD\_DEFAULT*

FMOD\_DEFAULT is a default sound type. Equivalent to all the defaults listed below. FMOD\_LOOP\_OFF, FMOD\_2D, FMOD\_HARDWARE.

### *FMOD\_LOOP\_OFF*

For non looping sounds. (DEFAULT). Overrides FMOD\_LOOP\_NORMAL / FMOD\_LOOP\_BIDI.

### *FMOD\_LOOP\_NORMAL*

For forward looping sounds.

### *FMOD\_LOOP\_BIDI*

For bidirectional looping sounds. (only works on software mixed static sounds).

### *FMOD\_2D*

Ignores any 3d processing. (DEFAULT).

### *FMOD\_3D*

Makes the sound positionable in 3D. Overrides FMOD\_2D.

### *FMOD\_HARDWARE*

Attempts to make sounds use hardware acceleration. (DEFAULT).

### *FMOD\_SOFTWARE*

Makes the sound be mixed by the FMOD CPU based software mixer. Overrides FMOD\_HARDWARE. Use this for FFT, DSP, compressed sample support, 2D multi-speaker support and other software related features.

### *FMOD\_CREATESTREAM*

Decompress at runtime, streaming from the source provided (ie from disk). Overrides FMOD\_CREATESAMPLE and FMOD\_CREATECOMPRESSED\_SAMPLE. Note a stream can only be played once at a time due to a stream

only having 1 stream buffer and file handle. Open multiple streams to have them play concurrently.

### *FMOD\_CREATEAMPLE*

Decompress at loadtime, decompressing or decoding whole file into memory as the target sample format (ie PCM). Fastest for FMOD\_SOFTWARE based playback and most flexible.

### *FMOD\_CREATECOMPRESSEDAMPLE*

Load MP2, MP3, IMAADPCM or XMA into memory and leave it compressed. During playback the FMOD software mixer will decode it in realtime as a 'compressed sample'. Can only be used in combination with FMOD\_SOFTWARE. Overrides FMOD\_CREATEAMPLE. If the sound data is not ADPCM, MPEG or XMA it will behave as if it was created with FMOD\_CREATEAMPLE and decode the sound into PCM.

### *FMOD\_OPENUSER*

Opens a user created static sample or stream. Use FMOD\_CREATEINDEXINFO to specify format and/or read callbacks. If a user created 'sample' is created with no read callback, the sample will be empty. Use Sound::lock and Sound::unlock to place sound data into the sound if this is the case.

### *FMOD\_OPENMEMORY*

"name\_or\_data" will be interpreted as a pointer to memory instead of filename for creating sounds. Use FMOD\_CREATEINDEXINFO to specify length. FMOD duplicates the memory into its own buffers. Can be freed after open.

### *FMOD\_OPENMEMORY\_POINT*

"name\_or\_data" will be interpreted as a pointer to memory instead of filename for creating sounds. Use FMOD\_CREATEINDEXINFO to specify length. This differs to FMOD\_OPENMEMORY in that it uses the memory as is, without duplicating the memory into its own buffers. FMOD\_SOFTWARE only. Doesn't work with FMOD\_HARDWARE, as sound hardware cannot access main ram on a lot of platforms. Cannot be freed after open, only after Sound::release. Will not work if the data is compressed and FMOD\_CREATECOMPRESSEDAMPLE is not used.

### *FMOD\_OPENRAW*

Will ignore file format and treat as raw pcm. Use FMOD\_CREATEINDEXINFO to specify format. Requires at least defaultfrequency, numchannels and format to be specified before it will open. Must be little endian data.

### *FMOD\_OPENONLY*

Just open the file, dont prebuffer or read. Good for fast opens for info, or when sound::readData is to be used.

### *FMOD\_ACCURATETIME*

For [System::createSound](#) - for accurate Sound::getLength/Channel::setPosition on VBR MP3, and MOD/S3M/XM/IT/MIDI files. Scans file first, so takes longer to open. FMOD\_OPENONLY does not affect this.

### *FMOD\_MPEGSEARCH*

For corrupted / bad MP3 files. This will search all the way through the file until it hits a valid MPEG header. Normally only searches for 4k.

### *FMOD\_NONBLOCKING*



For opening sounds and getting streamed subsounds (seeking) asynchronously. Use [Sound::getOpenState](#) to poll the state of the sound as it opens or retrieves the subsound in the background.

### *FMOD\_UNIQUE*

Unique sound, can only be played one at a time

### *FMOD\_3D\_HEADRELATIVE*

Make the sound's position, velocity and orientation relative to the listener.

### *FMOD\_3D\_WORLDRELATIVE*

Make the sound's position, velocity and orientation absolute (relative to the world). (DEFAULT)

### *FMOD\_3D\_LOGROLLOFF*

This sound will follow the standard logarithmic rolloff model where mindistance = full volume, maxdistance = where sound stops attenuating, and rolloff is fixed according to the global rolloff factor. (DEFAULT)

### *FMOD\_3D\_LINEARROLLOFF*

This sound will follow a linear rolloff model where mindistance = full volume, maxdistance = silence. Rolloffscale is ignored.

### *FMOD\_3D\_CUSTOMROLLOFF*

This sound will follow a rolloff model defined by [Sound::set3DCustomRolloff](#) / [Channel::set3DCustomRolloff](#).

### *FMOD\_3D\_IGNOREGEOMETRY*

Is not affect by geometry occlusion. If not specified in [Sound::setMode](#), or [Channel::setMode](#), the flag is cleared and it is affected by geometry again.

### *FMOD\_CDDA\_FORCEASPI*

For CDDA sounds only - use ASPI instead of NTSCSI to access the specified CD/DVD device.

### *FMOD\_CDDA\_JITTERCORRECT*

For CDDA sounds only - perform jitter correction. Jitter correction helps produce a more accurate CDDA stream at the cost of more CPU time.

### *FMOD\_UNICODE*

Filename is double-byte unicode.

### *FMOD\_IGNORETAGS*

Skips id3v2/asf/etc tag checks when opening a sound, to reduce seek/read overhead when opening files (helps with CD performance).

### *FMOD\_LOWMEM*

Removes some features from samples to give a lower memory overhead, like `Sound::getName`. See remarks.

### *FMOD\_LOADSECONDARYRAM*

Load sound into the secondary RAM of supported platform. On PS3, sounds will be loaded into RSX/VRAM.

### *FMOD\_VIRTUAL\_PLAYFROMSTART*

For sounds that start virtual (due to being quiet or low importance), instead of swapping back to audible, and playing at the correct offset according to time, this flag makes the sound play from the start.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_System::createSound](#)
- [\\_System::createStream](#)
- [\\_Sound::setMode](#)
- [\\_Sound::getMode](#)
- [\\_Channel::setMode](#)
- [\\_Channel::getMode](#)
- [\\_Sound::set3DCustomRolloff](#)
- [\\_Channel::set3DCustomRolloff](#)
- [\\_Sound::getOpenState](#)

# FMOD\_REVERB\_CHANNELFLAGS

Values for the Flags member of the [FMOD\\_REVERB\\_CHANNELPROPERTIES](#) structure. For EAX4 support with multiple reverb environments, set

FMOD\_REVERB\_CHANNELFLAGS\_ENVIRONMENT0, FMOD\_REVERB\_CHANNELFLAGS\_ENVIRONMENT1 or/and FMOD\_REVERB\_CHANNELFLAGS\_ENVIRONMENT2 in the flags member of

[FMOD\\_REVERB\\_CHANNELPROPERTIES](#) to specify which environment instance(s) to target.

Only up to 2 environments to target can be specified at once. Specifying three will result in an error. If the sound card does not support EAX4, the environment flag is ignored.

## Definition

```
#define FMOD_REVERB_CHANNELFLAGS_DIRECTHFAUTO 0x00000001
#define FMOD_REVERB_CHANNELFLAGS_ROOMAUTO 0x00000002
#define FMOD_REVERB_CHANNELFLAGS_ROOMHFAUTO 0x00000004
#define FMOD_REVERB_CHANNELFLAGS_ENVIRONMENT0 0x00000008
#define FMOD_REVERB_CHANNELFLAGS_ENVIRONMENT1 0x00000010
#define FMOD_REVERB_CHANNELFLAGS_ENVIRONMENT2 0x00000020
#define FMOD_REVERB_CHANNELFLAGS_ENVIRONMENT3 0x00000040
#define FMOD_REVERB_CHANNELFLAGS_ERROR (FMOD_REVERB_CHANNELFLAGS_DIRECTHFAUTO |
FMOD_REVERB_CHANNELFLAGS_ROOMAUTO | FMOD_REVERB_CHANNELFLAGS_ROOMHFAUTO |
FMOD_REVERB_CHANNELFLAGS_ENVIRONMENT0)
```

## Values

*FMOD\_REVERB\_CHANNELFLAGS\_DIRECTHFAUTO*

Automatic setting of 'Direct' due to distance from listener

*FMOD\_REVERB\_CHANNELFLAGS\_ROOMAUTO*

Automatic setting of 'Room' due to distance from listener

*FMOD\_REVERB\_CHANNELFLAGS\_ROOMHFAUTO*

Automatic setting of 'RoomHF' due to distance from listener

*FMOD\_REVERB\_CHANNELFLAGS\_ENVIRONMENT0*

EAX4/GameCube/Wii. Specify channel to target reverb instance 0.

*FMOD\_REVERB\_CHANNELFLAGS\_ENVIRONMENT1*

EAX4/GameCube/Wii. Specify channel to target reverb instance 1.

*FMOD\_REVERB\_CHANNELFLAGS\_ENVIRONMENT2*

EAX4/GameCube/Wii. Specify channel to target reverb instance 2.

*FMOD\_REVERB\_CHANNELFLAGS\_ENVIRONMENT3*

EAX5. Specify channel to target reverb instance 3.

*FMOD\_REVERB\_CHANNELFLAGS\_DEFAULT*

### **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

### **See Also**

- [FMOD\\_REVERB\\_CHANNELPROPERTIES](#)

# FMOD\_REVERB\_FLAGS

Values for the Flags member of the [FMOD\\_REVERB\\_PROPERTIES](#) structure.?

## Definition

```
#define FMOD_REVERB_FLAGS_DECAYTIMESCALE 0x00000001
#define FMOD_REVERB_FLAGS_REFLECTIONSSCALE 0x00000002
#define FMOD_REVERB_FLAGS_REFLECTIONSDELAYSCALE 0x00000004
#define FMOD_REVERB_FLAGS_REVERBSCALE 0x00000008
#define FMOD_REVERB_FLAGS_REVERBDELAYSCALE 0x00000010
#define FMOD_REVERB_FLAGS_DECAYHFLIMIT 0x00000020
#define FMOD_REVERB_FLAGS_ECHOTIMESCALE 0x00000040
#define FMOD_REVERB_FLAGS_ENVIRONMENTAL 0x00000080
#define FMOD_REVERB_FLAGS_COARSE0 0x00000100
#define FMOD_REVERB_FLAGS_COARSE1 0x00000200
#define FMOD_REVERB_FLAGS_HIGHPASS 0x00000400
#define FMOD_REVERB_FLAGS_HIGHPASS2 0x00000800
#define FMOD_REVERB_FLAGS_DEFAULT (FMOD_REVERB_FLAGS_DECAYTIMESCALE |
FMOD_REVERB_FLAGS_REFLECTIONSSCALE | FMOD_REVERB_FLAGS_REFLECTIONSDELAYSCALE |
FMOD_REVERB_FLAGS_REVERBSCALE | FMOD_REVERB_FLAGS_REVERBDELAYSCALE |
FMOD_REVERB_FLAGS_DECAYHFLIMIT | FMOD_REVERB_FLAGS_COARSE0 | FMOD_REVERB_FLAGS_COARSE1)
```

## Values

*FMOD\_REVERB\_FLAGS\_DECAYTIMESCALE*

'EnvSize' affects reverberation decay time

*FMOD\_REVERB\_FLAGS\_REFLECTIONSSCALE*

'EnvSize' affects reflection level

*FMOD\_REVERB\_FLAGS\_REFLECTIONSDELAYSCALE*

'EnvSize' affects initial reflection delay time

*FMOD\_REVERB\_FLAGS\_REVERBSCALE*

'EnvSize' affects reflections level

*FMOD\_REVERB\_FLAGS\_REVERBDELAYSCALE*

'EnvSize' affects late reverberation delay time

*FMOD\_REVERB\_FLAGS\_DECAYHFLIMIT*

AirAbsorptionHF affects DecayHFRatio

*FMOD\_REVERB\_FLAGS\_ECHOTIMESCALE*

'EnvSize' affects echo time

*FMOD\_REVERB\_FLAGS\_MODULATIONTIMESCALE*

'EnvSize' affects modulation time

*FMOD\_REVERB\_FLAGS\_CORE0*

PS2 Only - Reverb is applied to CORE0 (hw voices 0-23)

*FMOD\_REVERB\_FLAGS\_CORE1*

PS2 Only - Reverb is applied to CORE1 (hw voices 24-47)

*FMOD\_REVERB\_FLAGS\_HIGHQUALITYREVERB*

GameCube/Wii. Use high quality reverb

*FMOD\_REVERB\_FLAGS\_HIGHQUALITYDPL2REVERB*

GameCube/Wii. Use high quality DPL2 reverb

*FMOD\_REVERB\_FLAGS\_DEFAULT*

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [FMOD\\_REVERB\\_PROPERTIES](#)

# FMOD\_REVERB\_PRESETS

A set of predefined environment PARAMETERS, created by Creative Labs. These are used to initialize an FMOD\_REVERB\_PROPERTIES structure statically. ie FMOD\_REVERB\_PROPERTIES prop = FMOD\_PRESET\_GENERIC;

## Definition

```
#define FMOD_PRESET_GENERIC { 0, -1, 75f, 1.00f, -10000, -10000, 0, 1.00f, 1.00f, 1.0f, -2602, 0.00f, { 0.0f, 0.0f, 0.0f } , 200, 0.011f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 0.0f, 0.0f, 0.23f }
#define FMOD_PRESET_GENERIC { 0, 0, 75f, 1.00f, -1000, -100, 0, 1.0f, 0.8f, 1.0f, -2602, 0.00f, { 0.0f, 0.0f, 0.0f } , 200, 0.011f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 1, 1.4f, 1.00f, -1000, -6000, 0, 0.17f, 0.10f, 1.0f, -1204, 0.001f, { 0.0f, 0.0f, 0.0f } , 20, 0.002f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 2, 1.9f, 1.00f, -1000, -54, 0, 0.40f, 0.8f, 1.0f, -1646, 0.002f, { 0.0f, 0.0f, 0.0f } , 53, 0.03f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 3, 1.4f, 1.00f, -1000, -1200, 0, 1.0f, 0.54f, 1.0f, -370, 0.00f, { 0.0f, 0.0f, 0.0f } , 130, 0.011f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 60.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 4, 25f, 1.00f, -1000, -6000, 0, 0.50f, 0.10f, 1.0f, -1376, 0.03f, { 0.0f, 0.0f, 0.0f } , -1104, 0.004f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 5, 11.6f, 1.00f, -1000, -300, 0, 231f, 0.64f, 1.0f, -711, 0.012f, { 0.0f, 0.0f, 0.0f } , 8, 0.017f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 6, 2.6f, 1.00f, -1000, -76, 0, 432f, 0.59f, 1.0f, -78, 0.020f, { 0.0f, 0.0f, 0.0f } , -28, 0.30f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 7, 19.6f, 1.00f, -1000, -500, 0, 3.92f, 0.70f, 1.0f, -120, 0.020f, { 0.0f, 0.0f, 0.0f } , -2, 0.02f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 8, 14.6f, 1.00f, -1000, 0, 0, 2.91f, 1.30f, 1.0f, -602, 0.015f, { 0.0f, 0.0f, 0.0f } , -302, 0.022f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.1f }
#define FMOD_PRESET_GENERIC { 0, 9, 36.2f, 1.00f, -1000, -8, 0, 7.24f, 0.33f, 1.0f, -1166, 0.020f, { 0.0f, 0.0f, 0.0f } , 16, 0.30f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 10, 50.3f, 1.00f, -1000, -1000, 0, 10.5f, 0.2f, 1.0f, -602, 0.020f, { 0.0f, 0.0f, 0.0f } , 198, 0.30f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 11, 1.9f, 1.00f, -1000, -4000, 0, 0.30f, 0.10f, 1.0f, -181, 0.002f, { 0.0f, 0.0f, 0.0f } , -130, 0.30f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 12, 1.8f, 1.00f, -1000, -300, 0, 1.0f, 0.59f, 1.0f, -1219, 0.00f, { 0.0f, 0.0f, 0.0f } , 44, 0.011f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 13, 13.5f, 1.00f, -1000, -27, 0, 27.0f, 0.79f, 1.0f, -124, 0.013f, { 0.0f, 0.0f, 0.0f } , 395, 0.020f, { 0.0f, 0.0f, 0.0f } , 0.20f, 0.00f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 14, 75f, 0.30f, -1000, -20, 0, 1.0f, 0.86f, 1.0f, -1204, 0.00f, { 0.0f, 0.0f, 0.0f } , -4, 0.011f, { 0.0f, 0.0f, 0.0f } , 0.12f, 0.95f, 0.2f, 0.000f, -5.0f, 5000.0f, 20.0f, 0.0f, 100.0f, 100.0f, 0.2f }
#define FMOD_PRESET_GENERIC { 0, 15, 38.0f, 0.30f, -1000, -3300, 0, 1.0f, 0.54f, 1.0f, -260, 0.162f, { 0.0f, 0.0f, 0.0f } , -28, 0.088f, { 0.0f, 0.0f, 0.0f } ,
```

```

0.1 3 f, 1.00f, 0.3 f, 0.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 79.0f, 1 00.0f, 03 f}
# d f a MOD PRESE TCI T { 0, 16, 7 5 f, 05 0f, -1 000, - 800, 0, 1.4 f,
0.8 f, 1.0f, - 223 , 0.00 f,{ 0.0f,0.0f,0.0f} , -1 01 , 0.11 f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.3 f, 0.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 5 0.0f, 1 00.0f, 03 f}
# d f a MOD PRESE TMOUNAI N { 0, 17 , 1 00.0f, 0.2 f, -1 000, - 3 00, 0, 1.4 f,
0.2 f, 1.0f, - 2 80, 0.3 00f,{ 0.0f,0.0f,0.0f} , -1 4 4, 0.1 00f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 1.00f, 0.3 f, 0.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 2.0f, 1 00.0f, 01 f}
# d f a MOD PRESE TQUA RE { 0, 18, 17 5 f, 1.00f, -1 000, -1 000, 0, 1.4 f,
0.8 f, 1.0f, -1 0000, 0.06 f,{ 0.0f,0.0f,0.0f} , 5 00, 0.03 f,{ 0.0f,0.0f,0.0f} ,
0.1 3 f, 07 0f, 0.3 f, 0.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 1 00.0f, 1 00.0f, 03 f}
# d f a MOD PRESE T PAI N { 0, 19 , 425 f, 0.2 f, -1 000, - 2000, 0, 1.4 f,
05 0f, 1.0f, - 2466, 0.179 f,{ 0.0f,0.0f,0.0f} , -19 26, 0.1 00f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 1.00f, 0.3 f, 0.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 2.0f, 1 00.0f, 03 f}
# d f a MOD PRESE T AKI N D T { 0, 20, 83 f, 1.00f, -1 000, 0, 0, 1.6 f,
1 5 0f, 1.0f, -13 6 , 0.008f,{ 0.0f,0.0f,0.0f} , -1153 , 0.1 2f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.3 f, 0.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 1 00.0f, 1 00.0f, 01 f}
# d f a MOD PRESE TSEWE RE E { 0, 21 , 17 f, 0.80f, -1 000, -1 000, 0, 2.8 f,
0.1 4f, 1.0f, 42 , 0.1 4f,{ 0.0f,0.0f,0.0f} , 1 03 , 0.02 f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.3 f, 0.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 80.0f, 60.0f, 03 f}
# d f a MOD PRESE TUNE WA E R { 0, 22, 1.8f, 1.00f, -1 000, - 4000, 0, 1.4 f,
0.1 0f, 1.0f, - 44 , 0.00 f,{ 0.0f,0.0f,0.0f} , 17 00, 0.11 f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 1.18f, 0.3 48f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 1 00.0f, 1 00.0f, 03 f}
# d f a MOD PRESE T DRGGE D { 0, 23 , 1.9 f, 05 0f, -1 000, 0, 0, 839 f,
1 39 f, 1.0f, -115 , 0.002f,{ 0.0f,0.0f,0.0f} , 9 8 , 0.3 0f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.3 f, 1.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 1 00.0f, 1 00.0f, 01 f}
# d f a MOD PRESE T M Z Z { 0, 24, 1.8f, 0.60f, -1 000, - 400, 0, 17.3 f,
05 6f, 1.0f, -1713 , 0.020f,{ 0.0f,0.0f,0.0f} , - 613 , 0.3 0f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 1.00f, 0.8 f, 0.31 0f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 1 00.0f, 1 00.0f, 01 f}
# d f a MOD PRESE T BYC B TC { 0, 25 , 1.0f, 05 0f, -1 000, -151 , 0, 7 5 6f,
0.91 f, 1.0f, - 626, 0.020f,{ 0.0f,0.0f,0.0f} , 77 4, 0.3 0f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 4.00f, 1.000f, -5.0f, 5 000.0f, 3 0.0f, 0.0f, 1 00.0f, 1 00.0f, 01 f}
# d f a MOD PRESE T B 2 BOM { 0, 1, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 S TUM O A { 0, 2, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 S TUM O B { 0, 3, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 S TUM O C { 0, 4, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 A LL { 0, 5, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 S ACE { 0, 6, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 EC B { 0, 7, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.75 f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 E AY { 0, 8, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}
# d f a MOD PRESE T B 2 E E { 0, 9, 0, 0, 0, 0, 0, 0.0f,
0.0f, 0.0f, 0, 0.000f,{ 0.0f,0.0f,0.0f} , 0, 0.000f,{ 0.0f,0.0f,0.0f} ,
0.3 0f, 0.00f, 0.00f, 0.000f, 0.0f, 0000.0f, 0.0f, 0.0f, 0.0f, 0.0f, 03 1 f}

```

Values

*FMOD\_PRESET\_OFF*



*FMOD\_PRESET\_GENERIC*

*FMOD\_PRESET\_PADDEDCELL*

*FMOD\_PRESET\_ROOM*

*FMOD\_PRESET\_BATHROOM*

*FMOD\_PRESET\_LIVINGROOM*

*FMOD\_PRESET\_STONEROOM*

*FMOD\_PRESET\_AUDITORIUM*

*FMOD\_PRESET\_CONCERTHALL*

*FMOD\_PRESET\_CAVE*

*FMOD\_PRESET\_ARENA*

*FMOD\_PRESET\_HANGAR*

*FMOD\_PRESET\_CARPETTEDHALLWAY*

*FMOD\_PRESET\_HALLWAY*

*FMOD\_PRESET\_STONECORRIDOR*

*FMOD\_PRESET\_ALLEY*

*FMOD\_PRESET\_FOREST*

*FMOD\_PRESET\_CITY*

*FMOD\_PRESET\_MOUNTAINS*

*FMOD\_PRESET\_QUARRY*

*FMOD\_PRESET\_PLAIN*

*FMOD\_PRESET\_PARKINGLOT*

*FMOD\_PRESET\_SEWERPIPE*

*FMOD\_PRESET\_UNDERWATER*

*FMOD\_PRESET\_DRUGGED*

*FMOD\_PRESET\_DIZZY*

*FMOD\_PRESET\_PSYCHOTIC*

*FMOD\_PRESET\_PS2\_ROOM*

*FMOD\_PRESET\_PS2\_STUDIO\_A*

*FMOD\_PRESET\_PS2\_STUDIO\_B*

*FMOD\_PRESET\_PS2\_STUDIO\_C*

*FMOD\_PRESET\_PS2\_HALL*

*FMOD\_PRESET\_PS2\_SPACE*

*FMOD\_PRESET\_PS2\_ECHO*

*FMOD\_PRESET\_PS2\_DELAY*

*FMOD\_PRESET\_PS2\_PIPE*

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable,

PlayStation 3, Wii, Solaris

#### See Also

- [\\_System::setReverbProperties](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_TIMEUNIT

List of time types that can be returned by [Sound::getLength](#) and used with [Channel::setPosition](#) or [Channel::getPosition](#). FMOD\_TIMEUNIT\_SENTENCE\_MS, FMOD\_TIMEUNIT\_SENTENCE\_PCM, FMOD\_TIMEUNIT\_SENTENCE\_PCMBYTES, FMOD\_TIMEUNIT\_SENTENCE and FMOD\_TIMEUNIT\_SENTENCE\_SUBSOUND are only supported by Channel functions.

## Definition

```
#define FMOD_TIMEUNIT_MS 0x00000001
#define FMOD_TIMEUNIT_PCM 0x00000002
#define FMOD_TIMEUNIT_PCMBYTES 0x00000004
#define FMOD_TIMEUNIT_RAWBYTES 0x00000008
#define FMOD_TIMEUNIT_MODORDER 0x00000100
#define FMOD_TIMEUNIT_MODROW 0x00000200
#define FMOD_TIMEUNIT_MODDATERN 0x00000400
#define FMOD_TIMEUNIT_SENTENCE_MS 0x00010000
#define FMOD_TIMEUNIT_SENTENCE_PCM 0x00020000
#define FMOD_TIMEUNIT_SENTENCE_PCMBYTES 0x00040000
#define FMOD_TIMEUNIT_SENTENCE 0x00080000
#define FMOD_TIMEUNIT_SENTENCE_SUBSOUND 0x01000000
#define FMOD_TIMEUNIT_BUFFERED 0x10000000
```

## Values

*FMOD\_TIMEUNIT\_MS*

Milliseconds.

*FMOD\_TIMEUNIT\_PCM*

PCM Samples, related to milliseconds \* samplerate / 1000.

*FMOD\_TIMEUNIT\_PCMBYTES*

Bytes, related to PCM samples \* channels \* datawidth (ie 16bit = 2 bytes).

*FMOD\_TIMEUNIT\_RAWBYTES*

Raw file bytes of (compressed) sound data (does not include headers). Only used by [Sound::getLength](#) and [Channel::getPosition](#).

*FMOD\_TIMEUNIT\_MODORDER*

MOD/S3M/XM/IT. Order in a sequenced module format. Use [Sound::getFormat](#) to determine the PCM format being decoded to.

*FMOD\_TIMEUNIT\_MODROW*

MOD/S3M/XM/IT. Current row in a sequenced module format. [Sound::getLength](#) will return the number of rows in the currently playing or seeked to pattern.

### *FMOD\_TIMEUNIT\_MODPATTERN*

MOD/S3M/XM/IT. Current pattern in a sequenced module format. [Sound::getLength](#) will return the number of patterns in the song and [Channel::getPosition](#) will return the currently playing pattern.

### *FMOD\_TIMEUNIT\_SENTENCE\_MS*

Currently playing subsound in a sentence time in milliseconds.

### *FMOD\_TIMEUNIT\_SENTENCE\_PCM*

Currently playing subsound in a sentence time in PCM Samples, related to milliseconds \* samplerate / 1000.

### *FMOD\_TIMEUNIT\_SENTENCE\_PCMBYTES*

Currently playing subsound in a sentence time in bytes, related to PCM samples \* channels \* datawidth (ie 16bit = 2 bytes).

### *FMOD\_TIMEUNIT\_SENTENCE*

Currently playing sentence index according to the channel.

### *FMOD\_TIMEUNIT\_SENTENCE\_SUBSOUND*

Currently playing subsound index in a sentence.

### *FMOD\_TIMEUNIT\_BUFFERED*

Time value as seen by buffered stream. This is always ahead of audible time, and is only used for processing.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [Sound::getLength](#)
- [Channel::setPosition](#)
- [Channel::getPosition](#)

# Enumerations

[FMOD\\_CHANNELINDEX](#)  
[FMOD\\_CHANNEL\\_CALLBACKTYPE](#)  
[FMOD\\_DSP\\_CHORUS](#)  
[FMOD\\_DSP\\_COMPRESSOR](#)  
[FMOD\\_DSP\\_DISTORTION](#)  
[FMOD\\_DSP\\_ECHO](#)  
[FMOD\\_DSP\\_FFT\\_WINDOW](#)  
[FMOD\\_DSP\\_FLANGE](#)  
[FMOD\\_DSP\\_HIGHPASS](#)  
[FMOD\\_DSP\\_ITECHO](#)  
[FMOD\\_DSP\\_ITLOWPASS](#)  
[FMOD\\_DSP\\_LOWPASS](#)  
[FMOD\\_DSP\\_LOWPASS\\_SIMPLE](#)  
[FMOD\\_DSP\\_NORMALIZE](#)  
[FMOD\\_DSP\\_OSCILLATOR](#)  
[FMOD\\_DSP\\_PARAMEQ](#)  
[FMOD\\_DSP\\_PITCHSHIFT](#)  
[FMOD\\_DSP\\_RESAMPLER](#)  
[FMOD\\_DSP\\_REVERB](#)  
[FMOD\\_DSP\\_SFXREVERB](#)  
[FMOD\\_DSP\\_TYPE](#)  
[FMOD\\_OPENSTATE](#)  
[FMOD\\_OUTPUTTYPE](#)  
[FMOD\\_PLUGINTYPE](#)  
[FMOD\\_RESULT](#)  
[FMOD\\_SOUNDGROUP\\_BEHAVIOR](#)  
[FMOD\\_SOUND\\_FORMAT](#)  
[FMOD\\_SOUND\\_TYPE](#)  
[FMOD\\_SPEAKER](#)  
[FMOD\\_SPEAKERMAPTYPE](#)  
[FMOD\\_SPEAKERMODE](#)  
[FMOD\\_SYSTEM\\_CALLBACKTYPE](#)  
[FMOD\\_TAGDATATYPE](#)  
[FMOD\\_TAGTYPE](#)

# FMOD\_CHANNELINDEX

Special channel index values for FMOD functions.?

## Enumeration

```
typedef enum {  
    FMOD_CHANNEL_INDEX_FREE ,  
    FMOD_CHANNEL_INDEX_REUSE  
} FMOD_CHANNEL_INDEX
```

## Values

*FMOD\_CHANNEL\_FREE*

For a channel index, FMOD chooses a free voice using the priority system.

*FMOD\_CHANNEL\_REUSE*

For a channel index, re-use the channel handle that was passed in.

## Remarks

To get 'all' of the channels, use [System::getMasterChannelGroup](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::playSound](#)
- [System::playDSP](#)
- [System::getChannel](#)
- [System::getMasterChannelGroup](#)

# FMOD\_CHANNEL\_CALLBACKTYPE

These callback types are used with [Channel::setCallback](#)?

## Enumeration

```
typedef enum {
    FMOD_CHANNEL_CALLBACKTYPE_END,
    FMOD_CHANNEL_CALLBACKTYPE_VIRTUALVOICE,
    FMOD_CHANNEL_CALLBACKTYPE_SYNCPOINT,
    FMOD_CHANNEL_CALLBACKTYPE_MAX
} FMOD_CHANNEL_CALLBACKTYPE;
```

## Values

*FMOD\_CHANNEL\_CALLBACKTYPE\_END*

Called when a sound ends.

*FMOD\_CHANNEL\_CALLBACKTYPE\_VIRTUALVOICE*

Called when a voice is swapped out or swapped in.

*FMOD\_CHANNEL\_CALLBACKTYPE\_SYNCPOINT*

Called when a syncpoint is encountered. Can be from wav file markers.

*FMOD\_CHANNEL\_CALLBACKTYPE\_MAX*

Maximum number of callback types supported.

## Remarks

Each callback has commanddata parameters passed as int unique to the type of callback.

See reference to [FMOD\\_CHANNEL\\_CALLBACK](#) to determine what they might mean for each type of callback.

**Note!** Currently the user must call [System::update](#) for these callbacks to trigger!

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris



## See Also

- [\\_Channel::setCallback](#)
- [FMOD\\_CHANNEL\\_CALLBACK](#)
- [\\_System::update](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_DSP\_CHORUS

Parameter types for the [FMOD\\_DSP\\_TYPE\\_CHORUS](#) filter.?

## Enumeration

```
typedef enum {
    FMOD_DSP_PC_DRS_DRYMIX,
    FMOD_DSP_PC_DRS_WETMIX1,
    FMOD_DSP_PC_DRS_WETMIX2,
    FMOD_DSP_PC_DRS_WETMIX3,
    FMOD_DSP_PC_DRS_DELAY,
    FMOD_DSP_PC_DRS_RATE,
    FMOD_DSP_PC_DRS_DEPTH,
    FMOD_DSP_PC_DRS_FEEDBACK
} FMOD_DSP_PC_DRS;
```

## Values

*FMOD\_DSP\_CHORUS\_DRYMIX*

Volume of original signal to pass to output. 0.0 to 1.0. Default = 0.5.

*FMOD\_DSP\_CHORUS\_WETMIX1*

Volume of 1st chorus tap. 0.0 to 1.0. Default = 0.5.

*FMOD\_DSP\_CHORUS\_WETMIX2*

Volume of 2nd chorus tap. This tap is 90 degrees out of phase of the first tap. 0.0 to 1.0. Default = 0.5.

*FMOD\_DSP\_CHORUS\_WETMIX3*

Volume of 3rd chorus tap. This tap is 90 degrees out of phase of the second tap. 0.0 to 1.0. Default = 0.5.

*FMOD\_DSP\_CHORUS\_DELAY*

Chorus delay in ms. 0.1 to 100.0. Default = 40.0 ms.

*FMOD\_DSP\_CHORUS\_RATE*

Chorus modulation rate in hz. 0.0 to 20.0. Default = 0.8 hz.

*FMOD\_DSP\_CHORUS\_DEPTH*

Chorus modulation depth. 0.0 to 1.0. Default = 0.03.

*FMOD\_DSP\_CHORUS\_FEEDBACK*

Chorus feedback. Controls how much of the wet signal gets fed back into the chorus buffer. 0.0 to 1.0. Default = 0.0.

## Remarks

Chrous is an effect where the sound is more 'spacious' due to 1 to 3 versions of the sound being played along side the original signal but with the pitch of each copy modulating on a sine wave.

This is a highly configurable chorus unit. It supports 3 taps, small and large delay times and also feedback.

This unit also could be used to do a simple echo, or a flange effect.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)

# FMOD\_DSP\_COMPRESSOR

Parameter types for the [FMOD\\_DSP\\_TYPE\\_COMPRESSOR](#) unit. This is a simple linked multichannel software limiter that is uniform across the whole spectrum.

## Enumeration

```
typedef enum {
    FMOD_DSP_COMPRESSOR_THRESHOLD,
    FMOD_DSP_COMPRESSOR_ATTACK,
    FMOD_DSP_COMPRESSOR_RELEASE,
    FMOD_DSP_COMPRESSOR_GAINMAKEUP
}
```

## Values

*FMOD\_DSP\_COMPRESSOR\_THRESHOLD*

Threshold level (dB) in the range from -60 through 0. The default value is 0.

*FMOD\_DSP\_COMPRESSOR\_ATTACK*

Gain reduction attack time (milliseconds), in the range from 10 through 200. The default value is 50.

*FMOD\_DSP\_COMPRESSOR\_RELEASE*

Gain reduction release time (milliseconds), in the range from 20 through 1000. The default value is 50.

*FMOD\_DSP\_COMPRESSOR\_GAINMAKEUP*

Make-up gain (dB) applied after limiting, in the range from 0 through 30. The default value is 0.

## Remarks

The limiter is not guaranteed to catch every peak above the threshold level, because it cannot apply gain reduction instantaneously - the time delay is determined by the attack time. However setting the attack time too short will distort the sound, so it is a compromise. High level peaks can be avoided by using a short attack time - but not too short, and setting the threshold a few decibels below the critical level.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_DSP::SetParameter](#)
- [\\_DSP::GetParameter](#)
- [\\_FMODE\\_DSP\\_TYPE](#)
- [\\_System::addDSP](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_DSP\_DISTORTION

Parameter types for the [FMOD\\_DSP\\_TYPE\\_DISTORTION](#) filter.?

## Enumeration

```
typedef enum {  
    FMOD_DSP_DISTORTION_LEVEL,  
} FMOD_DSP_DISTORTION;
```

## Values

*FMOD\_DSP\_DISTORTION\_LEVEL*

Distortion value. 0.0 to 1.0. Default = 0.5.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)

# FMOD\_DSP\_ECHO

Parameter types for the [FMOD\\_DSP\\_TYPE\\_ECHO](#) filter.?

## Enumeration

```
typedef enum {
    FMOD_DSP_ECHO_DELAY,
    FMOD_DSP_ECHO_DECAYRATIO,
    FMOD_DSP_ECHO_MAXCHANNELS,
    FMOD_DSP_ECHO_DRYMIX,
    FMOD_DSP_ECHO_WETMIX
} FMOD_DSP_ECHO;
```

## Values

*FMOD\_DSP\_ECHO\_DELAY*

Echo delay in ms. 10 to 5000. Default = 500.

*FMOD\_DSP\_ECHO\_DECAYRATIO*

Echo decay per delay. 0 to 1. 1.0 = No decay, 0.0 = total decay (ie simple 1 line delay). Default = 0.5.

*FMOD\_DSP\_ECHO\_MAXCHANNELS*

Maximum channels supported. 0 to 16. 0 = same as fmod's default output polyphony, 1 = mono, 2 = stereo etc. See remarks for more. Default = 0. It is suggested to leave at 0!

*FMOD\_DSP\_ECHO\_DRYMIX*

Volume of original signal to pass to output. 0.0 to 1.0. Default = 1.0.

*FMOD\_DSP\_ECHO\_WETMIX*

Volume of echo signal to pass to output. 0.0 to 1.0. Default = 1.0.

## Remarks

Note. Every time the delay is changed, the plugin re-allocates the echo buffer. This means the echo will dissappear at that time while it refills its new buffer.

Larger echo delays result in larger amounts of memory allocated.

'maxchannels' also dictates the amount of memory allocated. By default, the maxchannels value is 0. If FMOD is set to stereo, the echo unit will allocate enough memory for 2 channels. If it is 5.1, it will allocate enough memory for a 6 channel echo, etc.

If the echo effect is only ever applied to the global mix (ie it was added with System::addDSP), then 0 is the value to set as it will be enough to handle all speaker modes.

When the echo is added to a channel (ie Channel::addDSP) then the channel count that comes in could be anything

from 1 to 8 possibly. It is only in this case where you might want to increase the channel count above the output's channel count.

If a channel echo is set to a lower number than the sound's channel count that is coming in, it will not echo the sound.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## **See Also**

- [\\_DSP::setParameter](#)
- [\\_DSP::getParameter](#)
- [\\_FMOD\\_DSP\\_TYPE](#)



# FMOD\_DSP\_FFT\_WINDOW

List of windowing methods used in spectrum analysis to reduce leakage / transient signals interfering with the analysis.  
?This is a problem with analysis of continuous signals that only have a small portion of the signal sample (the fft window size).

?Windowing the signal with a curve or triangle tapers the sides of the fft window to help alleviate this problem.?

## Enumeration

```
typedef enum {  
    FMOD_DSP_FFT_WINDOW_RECT,  
    FMOD_DSP_FFT_WINDOW_TRIANGLE,  
    FMOD_DSP_FFT_WINDOW_HAMMING,  
    FMOD_DSP_FFT_WINDOW_HANNING,  
    FMOD_DSP_FFT_WINDOW_BLACKMAN,  
    FMOD_DSP_FFT_WINDOW_BLACKMANHARRIS,  
    FMOD_DSP_FFT_WINDOW_MAX  
}
```

## Values

*FMOD\_DSP\_FFT\_WINDOW\_RECT*

$w[n] = 1.0$

*FMOD\_DSP\_FFT\_WINDOW\_TRIANGLE*

$w[n] = \text{TRI}(2n/N)$

*FMOD\_DSP\_FFT\_WINDOW\_HAMMING*

$w[n] = 0.54 - (0.46 * \cos(n/N))$

*FMOD\_DSP\_FFT\_WINDOW\_HANNING*

$w[n] = 0.5 * (1.0 - \cos(n/N))$

*FMOD\_DSP\_FFT\_WINDOW\_BLACKMAN*

$w[n] = 0.42 - (0.5 * \cos(n/N)) + (0.08 * \cos(2.0 * n/N))$

*FMOD\_DSP\_FFT\_WINDOW\_BLACKMANHARRIS*

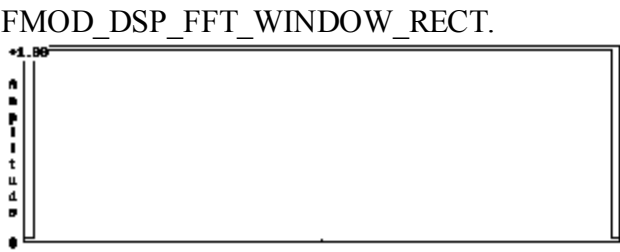
$w[n] = 0.35875 - (0.48829 * \cos(1.0 * n/N)) + (0.14128 * \cos(2.0 * n/N)) - (0.01168 * \cos(3.0 * n/N))$

*FMOD\_DSP\_FFT\_WINDOW\_MAX*

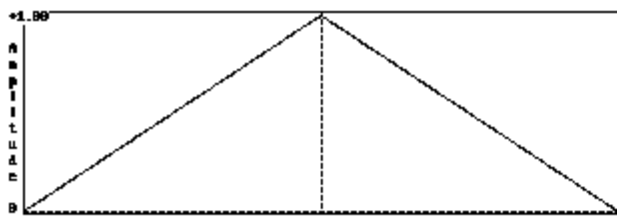
Maximum number of FFT window types supported.

## Remarks

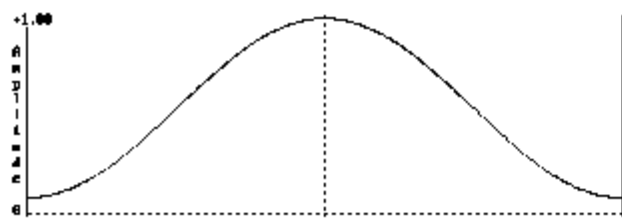
Cyclic signals such as a sine wave that repeat their cycle in a multiple of the window size do not need windowing. I.e. If the sine wave repeats every 1024, 512, 256 etc samples and the FMOD fft window is 1024, then the signal would not need windowing. Not windowing is the same as [FMOD\\_DSP\\_FFT\\_WINDOW\\_RECT](#), which is the default. If the cycle of the signal (ie the sine wave) is not a multiple of the window size, it will cause frequency abnormalities, so a different windowing method is needed.



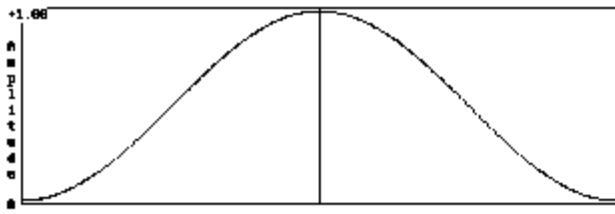
FMOD\_DSP\_FFT\_WINDOW\_TRIANGLE.



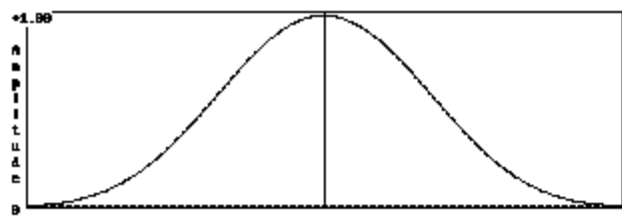
FMOD\_DSP\_FFT\_WINDOW\_HAMMING.



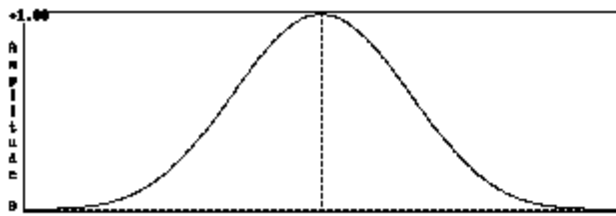
FMOD\_DSP\_FFT\_WINDOW\_HANNING.



FMOD\_DSP\_FFT\_WINDOW\_BLACKMAN.



FMOD\_DSP\_FFT\_WINDOW\_BLACKMANHARRIS.



## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getSpectrum](#)
- [Channel::getSpectrum](#)





# FMOD\_DSP\_FLANGE

Parameter types for the [FMOD\\_DSP\\_TYPE\\_FLANGE](#) filter.?

## Enumeration

```
typedef enum {
    FMOD_DSP_FLANGE_DRYMIX,
    FMOD_DSP_FLANGE_WETMIX,
    FMOD_DSP_FLANGE_DEPTH,
    FMOD_DSP_FLANGE_RATE
} FMOD_DSP_FLANGE;
```

## Values

*FMOD\_DSP\_FLANGE\_DRYMIX*

Volume of original signal to pass to output. 0.0 to 1.0. Default = 0.45.

*FMOD\_DSP\_FLANGE\_WETMIX*

Volume of flange signal to pass to output. 0.0 to 1.0. Default = 0.55.

*FMOD\_DSP\_FLANGE\_DEPTH*

Flange depth. 0.01 to 1.0. Default = 1.0.

*FMOD\_DSP\_FLANGE\_RATE*

Flange speed in hz. 0.0 to 20.0. Default = 0.1.

## Remarks

Flange is an effect where the signal is played twice at the same time, and one copy slides back and forth creating a whooshing or flanging effect.

As there are 2 copies of the same signal, by default each signal is given 50% mix, so that the total is not louder than the original unaffected signal.

Flange depth is a percentage of a 10ms shift from the original signal. Anything above 10ms is not considered flange because to the ear it begins to 'echo' so 10ms is the highest value possible.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_DSP::setParameter](#)
- [\\_DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_DSP\_HIGHPASS

Parameter types for the [FMOD\\_DSP\\_TYPE\\_HIGHPASS](#) filter.?

## Enumeration

```
typedef enum {  
    FMOD_DSP_HIGHPASS_CUTOFF,  
    FMOD_DSP_HIGHPASS_RESONANCE  
} FMOD_DSP_HIGHPASS;
```

## Values

*FMOD\_DSP\_HIGHPASS\_CUTOFF*

Highpass cutoff frequency in hz. 1.0 to output 22000.0. Default = 5000.0.

*FMOD\_DSP\_HIGHPASS\_RESONANCE*

Highpass resonance Q value. 1.0 to 10.0. Default = 1.0.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)

# FMOD\_DSP\_ITECHO

Parameter types for the [FMOD\\_DSP\\_TYPE\\_ITECHO](#) filter.

?This is effectively a software based echo filter that emulates the DirectX DMO echo effect. Impulse tracker files can support this, and FMOD will produce the effect on ANY platform, not just those that support DirectX effects!  
?

## Enumeration

```
typedef enum {
    FMOD_DSP_ITECHO_WETDRY MIX,
    FMOD_DSP_ITECHO_FEEDBACK,
    FMOD_DSP_ITECHO_LEFTDELAY,
    FMOD_DSP_ITECHO_RIGHTDELAY,
    FMOD_DSP_ITECHO_PANDELAY
} FMOD_DSP_ITECHO;
```

## Values

*FMOD\_DSP\_ITECHO\_WETDRY MIX*

Ratio of wet (processed) signal to dry (unprocessed) signal. Must be in the range from 0.0 through 100.0 (all wet). The default value is 50.

*FMOD\_DSP\_ITECHO\_FEEDBACK*

Percentage of output fed back into input, in the range from 0.0 through 100.0. The default value is 50.

*FMOD\_DSP\_ITECHO\_LEFTDELAY*

Delay for left channel, in milliseconds, in the range from 1.0 through 2000.0. The default value is 500 ms.

*FMOD\_DSP\_ITECHO\_RIGHTDELAY*

Delay for right channel, in milliseconds, in the range from 1.0 through 2000.0. The default value is 500 ms.

*FMOD\_DSP\_ITECHO\_PANDELAY*

Value that specifies whether to swap left and right delays with each successive echo. The default value is zero, meaning no swap. Possible values are defined as 0.0 (equivalent to FALSE) and 1.0 (equivalent to TRUE).  
CURRENTLY NOT SUPPORTED.

## Remarks

Note. Every time the delay is changed, the plugin re-allocates the echo buffer. This means the echo will disappear at that time while it refills its new buffer.

Larger echo delays result in larger amounts of memory allocated.

As this is a stereo filter made mainly for IT playback, it is targeted for stereo signals.

With mono signals only the [FMOD\\_DSP\\_ITECHO\\_LEFTDELAY](#) is used.  
For multichannel signals (>2) there will be no echo on those channels.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::SetParameter](#)
- [DSP::GetParameter](#)
- [FMOD\\_DSP\\_TYPE](#)
- [System::addDSP](#)

# FMOD\_DSP\_ITLOWPASS

Parameter types for the [FMOD\\_DSP\\_TYPE\\_ITLOWPASS](#) filter.

?This is different to the default [FMOD\\_DSP\\_TYPE\\_ITLOWPASS](#) filter in that it uses a different quality algorithm and is?the filter used to produce the correct sounding playback in .IT files.

?FMOD Ex's .IT playback uses this filter.

?

## Enumeration

```
typedef enum {  
    FMOD_DSP_ITLOWPASS_CUTOFF,  
    FMOD_DSP_ITLOWPASS_RESONANCE  
} FMOD_DSP_ITLOWPASS;
```

## Values

*FMOD\_DSP\_ITLOWPASS\_CUTOFF*

Lowpass cutoff frequency in hz. 1.0 to 22000.0. Default = 5000.0/

*FMOD\_DSP\_ITLOWPASS\_RESONANCE*

Lowpass resonance Q value. 0.0 to 127.0. Default = 1.0.

## Remarks

Note! This filter actually has a limited cutoff frequency below the specified maximum, due to its limited design, so for a more open range filter use [FMOD\\_DSP\\_LOWPASS](#) or if you don't mind not having resonance, [FMOD\\_DSP\\_LOWPASS\\_SIMPLE](#).

The effective maximum cutoff is about 8060hz.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)



# FMOD\_DSP\_LOWPASS

Parameter types for the [FMOD\\_DSP\\_TYPE\\_LOWPASS](#) filter.?

## Enumeration

```
typedef enum {  
    FMOD_DSP_LOWPASS_CUTOFF,  
    FMOD_DSP_LOWPASS_RESONANCE  
} FMOD_DSP_LOWPASS;
```

## Values

*FMOD\_DSP\_LOWPASS\_CUTOFF*

Lowpass cutoff frequency in hz. 10.0 to 22000.0. Default = 5000.0.

*FMOD\_DSP\_LOWPASS\_RESONANCE*

Lowpass resonance Q value. 1.0 to 10.0. Default = 1.0.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)



# FMOD\_DSP\_LOWPASS\_SIMPLE

Parameter types for the [FMOD\\_DSP\\_TYPE\\_LOWPASS\\_SIMPLE](#) filter.

?This is a very simple low pass filter, based on two single-pole RC time-constant modules.?The emphasis is on speed rather than accuracy, so this should not be used for task requiring critical filtering.

?

## Enumeration

```
typedef enum {  
    FMOD_DSP_LOWPASS_SIMPLE_CUTOFF  
} FMOD_DSP_LOWPASS_SIMPLE;
```

## Values

*FMOD\_DSP\_LOWPASS\_SIMPLE\_CUTOFF*

Lowpass cutoff frequency in hz. 10.0 to 22000.0. Default = 5000.0

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)

# FMOD\_DSP\_NORMALIZE

Parameter types for the [FMOD\\_DSP\\_TYPE\\_NORMALIZE](#) filter.?

## Enumeration

```
typedef enum {  
    FMOD_DSP_NORMALIZE_FADETIME,  
    FMOD_DSP_NORMALIZE_THRESHOLD,  
    FMOD_DSP_NORMALIZE_MAXAMP  
} FMOD_DSP_NORMALIZE;
```

## Values

*FMOD\_DSP\_NORMALIZE\_FADETIME*

Time to ramp the silence to full in ms. 0.0 to 20000.0. Default = 5000.0.

*FMOD\_DSP\_NORMALIZE\_THRESHOLD*

Lower volume range threshold to ignore. 0.0 to 1.0. Default = 0.1. Raise higher to stop amplification of very quiet signals.

*FMOD\_DSP\_NORMALIZE\_MAXAMP*

Maximum amplification allowed. 1.0 to 100000.0. Default = 20.0. 1.0 = no amplification, higher values allow more boost.

## Remarks

Normalize amplifies the sound based on the maximum peaks within the signal.

For example if the maximum peaks in the signal were 50% of the bandwidth, it would scale the whole sound by 2.

The lower threshold value makes the normalizer ignores peaks below a certain point, to avoid over-amplification if a loud signal suddenly came in, and also to avoid amplifying to maximum things like background hiss.

Because FMOD is a realtime audio processor, it doesn't have the luxury of knowing the peak for the whole sound (ie it can't see into the future), so it has to process data as it comes in.

To avoid very sudden changes in volume level based on small samples of new data, fmod fades towards the desired amplification which makes for smooth gain control. The fadetime parameter can control this.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_DSP::setParameter](#)
- [\\_DSP::getParameter](#)
- [\\_FMODE\\_DSP\\_TYPE](#)

# FMOD\_DSP\_OSCILLATOR

Parameter types for the [FMOD\\_DSP\\_TYPE\\_OSCILLATOR](#) filter.?

## Enumeration

```
typedef enum {
    FMOD_DSP_OSCILLATOR_SINE,
    FMOD_DSP_OSCILLATOR_SQUARE,
    FMOD_DSP_OSCILLATOR_SAWUP,
    FMOD_DSP_OSCILLATOR_SAWDOWN,
    FMOD_DSP_OSCILLATOR_TRIANGLE,
    FMOD_DSP_OSCILLATOR_NOISE
}
```

## Values

*FMOD\_DSP\_OSCILLATOR\_TYPE*

Waveform type. 0 = sine. 1 = square. 2 = sawup. 3 = sawdown. 4 = triangle. 5 = noise.

*FMOD\_DSP\_OSCILLATOR\_RATE*

Frequency of the sinewave in hz. 1.0 to 22000.0. Default = 220.0.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)

# FMOD\_DSP\_PARAMEQ

Parameter types for the [FMOD\\_DSP\\_TYPE\\_PARAMEQ](#) filter.?

## Enumeration

```
typedef enum {  
    FMOD_DSP_PARAMEQ_CENTER,  
    FMOD_DSP_PARAMEQ_BANDWIDTH,  
    FMOD_DSP_PARAMEQ_GAIN  
} FMOD_DSP_PARAMEQ;
```

## Values

*FMOD\_DSP\_PARAMEQ\_CENTER*

Frequency center. 20.0 to 22000.0. Default = 8000.0.

*FMOD\_DSP\_PARAMEQ\_BANDWIDTH*

Octave range around the center frequency to filter. 0.2 to 5.0. Default = 1.0.

*FMOD\_DSP\_PARAMEQ\_GAIN*

Frequency Gain. 0.05 to 3.0. Default = 1.0.

## Remarks

Parametric EQ is a bandpass filter that attenuates or amplifies a selected frequency and its neighbouring frequencies.

To create a multi-band EQ create multiple [FMOD\\_DSP\\_TYPE\\_PARAMEQ](#) units and set each unit to different frequencies, for example 1000hz, 2000hz, 4000hz, 8000hz, 16000hz with a range of 1 octave each.

When a frequency has its gain set to 1.0, the sound will be unaffected and represents the original signal exactly.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [DSP::setParameter](#)
- [DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)



# FMOD\_DSP\_PITCHSHIFT

Parameter types for the [FMOD\\_DSP\\_TYPE\\_PITCHSHIFT](#) filter.?

## Enumeration

```
typedef enum {
    FMOD_DSP_PITCHSHIFT_FFT_FFT_H,
    FMOD_DSP_PITCHSHIFT_FFT_FFTBI_T,
    FMOD_DSP_PITCHSHIFT_FFT_OVERLAP,
    FMOD_DSP_PITCHSHIFT_FFT_MAXCHANNELS,
} FMOD_DSP_PITCHSHIFT_FFT;
```

## Values

*FMOD\_DSP\_PITCHSHIFT\_PITCH*

Pitch value. 0.5 to 2.0. Default = 1.0. 0.5 = one octave down, 2.0 = one octave up. 1.0 does not change the pitch.

*FMOD\_DSP\_PITCHSHIFT\_FFTSIZE*

FFT window size. 256, 512, 1024, 2048, 4096. Default = 1024. Increase this to reduce 'smearing'. This effect is a warbling sound similar to when an mp3 is encoded at very low bitrates.

*FMOD\_DSP\_PITCHSHIFT\_OVERLAP*

Removed. Do not use. FMOD now uses 4 overlaps and cannot be changed.

*FMOD\_DSP\_PITCHSHIFT\_MAXCHANNELS*

Maximum channels supported. 0 to 16. 0 = same as fmod's default output polyphony, 1 = mono, 2 = stereo etc. See remarks for more. Default = 0. It is suggested to leave at 0!

## Remarks

This pitch shifting unit can be used to change the pitch of a sound without speeding it up or slowing it down. It can also be used for time stretching or scaling, for example if the pitch was doubled, and the frequency of the sound was halved, the pitch of the sound would sound correct but it would be twice as slow.

**Warning!** This filter is very computationally expensive! Similar to a vocoder, it requires several overlapping FFT and IFFT's to produce smooth output, and can require around 440mhz for 1 stereo 48khz signal using the default settings. Reducing the signal to mono will half the cpu usage.

Reducing this will lower audio quality, but what settings to use are largely dependant on the sound being played. A noisy polyphonic signal will need higher fft size compared to a speaking voice for example.

This pitch shifter is based on the pitch shifter code at <http://www.dspdimension.com>, written by Stephan M. Bernsee. The original code is COPYRIGHT 1999-2003 Stephan M. Bernsee .

'*maxchannels*' dictates the amount of memory allocated. By default, the maxchannels value is 0. If FMOD is set to

stereo, the pitch shift unit will allocate enough memory for 2 channels. If it is 5.1, it will allocate enough memory for a 6 channel pitch shift, etc.

If the pitch shift effect is only ever applied to the global mix (ie it was added with `System::addDSP`), then 0 is the value to set as it will be enough to handle all speaker modes.

When the pitch shift is added to a channel (ie `Channel::addDSP`) then the channel count that comes in could be anything from 1 to 8 possibly. It is only in this case where you might want to increase the channel count above the output's channel count.

If a channel pitch shift is set to a lower number than the sound's channel count that is coming in, it will not pitch shift the sound.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_DSP::setParameter](#)
- [\\_DSP::getParameter](#)
- [\\_FMOD\\_DSP\\_TYPE](#)



# FMOD\_DSP\_RESAMPLER

List of interpolation types that the FMOD Ex software mixer supports.?

## Enumeration

```
typedef enum {
    FMOD_DSP_RESAMPLER_NOINTERP,
    FMOD_DSP_RESAMPLER_LINEAR,
    FMOD_DSP_RESAMPLER_CUBIC,
    FMOD_DSP_RESAMPLER_SPLINE,
    FMOD_DSP_RESAMPLER_MAX
}
```

## Values

*FMOD\_DSP\_RESAMPLER\_NOINTERP*

No interpolation. High frequency aliasing hiss will be audible depending on the sample rate of the sound.

*FMOD\_DSP\_RESAMPLER\_LINEAR*

Linear interpolation (default method). Fast and good quality, causes very slight lowpass effect on low frequency sounds.

*FMOD\_DSP\_RESAMPLER\_CUBIC*

Cubic interpolation. Slower than linear interpolation but better quality.

*FMOD\_DSP\_RESAMPLER\_SPLINE*

5 point spline interpolation. Slowest resampling method but best quality.

*FMOD\_DSP\_RESAMPLER\_MAX*

Maximum number of resample methods supported.

## Remarks

The default resampler type is [FMOD\\_DSP\\_RESAMPLER\\_LINEAR](#).

Use [System::setSoftwareFormat](#) to tell FMOD the resampling quality you require for [FMOD\\_SOFTWARE](#) based sounds.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

**See Also**

- [System:setSoftwareFormat](#)
- [System:getSoftwareFormat](#)

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# FMOD\_DSP\_REVERB

Parameter types for the [FMOD\\_DSP\\_TYPE\\_REVERB](#) filter.?

## Enumeration

```
typedef enum {
    FMOD_DSP_REVERB_ROOMSIZE,
    FMOD_DSP_REVERB_DAMP,
    FMOD_DSP_REVERB_WETMIX,
    FMOD_DSP_REVERB_DRYMIX,
    FMOD_DSP_REVERB_WIDTH,
    FMOD_DSP_REVERB_MODE
}
```

## Values

*FMOD\_DSP\_REVERB\_ROOMSIZE*

Roomsize. 0.0 to 1.0. Default = 0.5

*FMOD\_DSP\_REVERB\_DAMP*

Damp. 0.0 to 1.0. Default = 0.5

*FMOD\_DSP\_REVERB\_WETMIX*

Wet mix. 0.0 to 1.0. Default = 0.33

*FMOD\_DSP\_REVERB\_DRYMIX*

Dry mix. 0.0 to 1.0. Default = 0.66

*FMOD\_DSP\_REVERB\_WIDTH*

Stereo width. 0.0 to 1.0. Default = 1.0

*FMOD\_DSP\_REVERB\_MODE*

Mode. 0 (normal), 1 (freeze). Default = 0

## Remarks

Based on freeverb by Jezar at Dreampoint - <http://www.dreampoint.co.uk>.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

#### See Also

- [\\_DSP::setParameter](#)
- [\\_DSP::getParameter](#)
- [FMOD\\_DSP\\_TYPE](#)

# FMOD\_DSP\_SFXREVERB

Parameter types for the [FMOD\\_DSP\\_TYPE\\_SFXREVERB](#) unit.

?

## Enumeration

```
typedef enum {
    FMOD_DSP_SFXREVERB_DRYLEVEL,
    FMOD_DSP_SFXREVERB_ROOM,
    FMOD_DSP_SFXREVERB_ROOMHF,
    FMOD_DSP_SFXREVERB_ROOMROLLOFFFACTOR,
    FMOD_DSP_SFXREVERB_DECAYTIME,
    FMOD_DSP_SFXREVERB_DECAYHFRATIO,
    FMOD_DSP_SFXREVERB_REVERBTOSE,
    FMOD_DSP_SFXREVERB_REVERBTOSEAY,
    FMOD_DSP_SFXREVERB_REVERB,
    FMOD_DSP_SFXREVERB_REVERBAY,
    FMOD_DSP_SFXREVERB_WFUSION,
    FMOD_DSP_SFXREVERB_ENIT,
    FMOD_DSP_SFXREVERB_HFEENE,
    FMOD_DSP_SFXREVERB_ROOMLF,
    FMOD_DSP_SFXREVERB_LFEENE,
} FMOD_DSP_SFXREVERB;
```

## Values

*FMOD\_DSP\_SFXREVERB\_DRYLEVEL*

Dry Level : Mix level of dry signal in output in mB. Ranges from -10000.0 to 0.0. Default is 0.

*FMOD\_DSP\_SFXREVERB\_ROOM*

Room : Room effect level at low frequencies in mB. Ranges from -10000.0 to 0.0. Default is 0.0.

*FMOD\_DSP\_SFXREVERB\_ROOMHF*

Room HF : Room effect high-frequency level re. low frequency level in mB. Ranges from -10000.0 to 0.0. Default is 0.0.

*FMOD\_DSP\_SFXREVERB\_ROOMROLLOFFFACTOR*

Room Rolloff : Like DS3D flRolloffFactor but for room effect. Ranges from 0.0 to 10.0. Default is 10.0

*FMOD\_DSP\_SFXREVERB\_DECAYTIME*

Decay Time : Reverberation decay time at low-frequencies in seconds. Ranges from 0.1 to 20.0. Default is 1.0.

*FMOD\_DSP\_SFXREVERB\_DECAYHFRATIO*

Decay HF Ratio : High-frequency to low-frequency decay time ratio. Ranges from 0.1 to 2.0. Default is 0.5.

### *FMOD\_DSP\_SFXREVERB\_REFLECTIONSLEVEL*

Reflections : Early reflections level relative to room effect in mB. Ranges from -10000.0 to 1000.0. Default is -10000.0.

### *FMOD\_DSP\_SFXREVERB\_REFLECTIONSDELAY*

Reflect Delay : Delay time of first reflection in seconds. Ranges from 0.0 to 0.3. Default is 0.02.

### *FMOD\_DSP\_SFXREVERB\_REVERBLEVEL*

Reverb : Late reverberation level relative to room effect in mB. Ranges from -10000.0 to 2000.0. Default is 0.0.

### *FMOD\_DSP\_SFXREVERB\_REVERBDELAY*

Reverb Delay : Late reverberation delay time relative to first reflection in seconds. Ranges from 0.0 to 0.1. Default is 0.04.

### *FMOD\_DSP\_SFXREVERB\_DIFFUSION*

Diffusion : Reverberation diffusion (echo density) in percent. Ranges from 0.0 to 100.0. Default is 100.0.

### *FMOD\_DSP\_SFXREVERB\_DENSITY*

Density : Reverberation density (modal density) in percent. Ranges from 0.0 to 100.0. Default is 100.0.

### *FMOD\_DSP\_SFXREVERB\_HFREFERENCE*

HF Reference : Reference high frequency in Hz. Ranges from 20.0 to 20000.0. Default is 5000.0.

### *FMOD\_DSP\_SFXREVERB\_ROOMLF*

Room LF : Room effect low-frequency level in mB. Ranges from -10000.0 to 0.0. Default is 0.0.

### *FMOD\_DSP\_SFXREVERB\_LFREFERENCE*

LF Reference : Reference low-frequency in Hz. Ranges from 20.0 to 1000.0. Default is 250.0.

## **Remarks**

This is a high quality I3DL2 based reverb which improves greatly on [FMOD\\_DSP\\_REVERB](#). On top of the I3DL2 property set, "Dry Level" is also included to allow the dry mix to be changed.

Currently [FMOD\\_DSP\\_SFXREVERB\\_REFLECTIONSLEVEL](#), [FMOD\\_DSP\\_SFXREVERB\\_REFLECTIONSDELAY](#) and [FMOD\\_DSP\\_SFXREVERB\\_REVERBDELAY](#) are not enabled but will come in future versions.

These properties can be set with presets in [FMOD\\_REVERB\\_PRESETS](#).

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable,

PlayStation 3, Wii

#### See Also

- [DSP::SetParameter](#)
- [DSP::GetParameter](#)
- [FMOD\\_DSP\\_TYPE](#)
- [System::addDSP](#)
- [FMOD\\_REVERB\\_PRESETS](#)

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# FMOD\_DSP\_TYPE

These definitions can be used for creating FMOD defined special effects or DSP units.?

## Enumeration

```
typedef enum {
    FMOD_DSP_TYPE_UNKNOWN,
    FMOD_DSP_TYPE_MIXER,
    FMOD_DSP_TYPE_OSCILLATOR,
    FMOD_DSP_TYPE_LOWPASS,
    FMOD_DSP_TYPE_ITLOWPASS,
    FMOD_DSP_TYPE_HIGHPASS,
    FMOD_DSP_TYPE_ECORE,
    FMOD_DSP_TYPE_FADER,
    FMOD_DSP_TYPE_DSDRTO,
    FMOD_DSP_TYPE_NMALE,
    FMOD_DSP_TYPE_AAMEQ,
    FMOD_DSP_TYPE_PITCHBEND,
    FMOD_DSP_TYPE_CHORUS,
    FMOD_DSP_TYPE_ECHO,
    FMOD_DSP_TYPE_STEREO,
    FMOD_DSP_TYPE_WIDEN,
    FMOD_DSP_TYPE_I_TCHORUS,
    FMOD_DSP_TYPE_COMPRESSOR,
    FMOD_DSP_TYPE_SFXECHO,
    FMOD_DSP_TYPE_LOWPASS_SIMP
}
```

## Values

*FMOD\_DSP\_TYPE\_UNKNOWN*

This unit was created via a non FMOD plugin so has an unknown purpose.

*FMOD\_DSP\_TYPE\_MIXER*

This unit does nothing but take inputs and mix them together then feed the result to the soundcard unit.

*FMOD\_DSP\_TYPE\_OSCILLATOR*

This unit generates sine/square/saw/triangle or noise tones.

*FMOD\_DSP\_TYPE\_LOWPASS*

This unit filters sound using a high quality, resonant lowpass filter algorithm but consumes more CPU time.

*FMOD\_DSP\_TYPE\_ITLOWPASS*

This unit filters sound using a resonant lowpass filter algorithm that is used in Impulse Tracker, but with limited cutoff range (0 to 8060hz).

*FMOD\_DSP\_TYPE\_HIGHPASS*



This unit filters sound using a resonant highpass filter algorithm.

#### *FMOD\_DSP\_TYPE\_ECHO*

This unit produces an echo on the sound and fades out at the desired rate.

#### *FMOD\_DSP\_TYPE\_FLANGE*

This unit produces a flange effect on the sound.

#### *FMOD\_DSP\_TYPE\_DISTORTION*

This unit distorts the sound.

#### *FMOD\_DSP\_TYPE\_NORMALIZE*

This unit normalizes or amplifies the sound to a certain level.

#### *FMOD\_DSP\_TYPE\_PARAMEQ*

This unit attenuates or amplifies a selected frequency range.

#### *FMOD\_DSP\_TYPE\_PITCHSHIFT*

This unit bends the pitch of a sound without changing the speed of playback.

#### *FMOD\_DSP\_TYPE\_CHORUS*

This unit produces a chorus effect on the sound.

#### *FMOD\_DSP\_TYPE\_REVERB*

This unit produces a reverb effect on the sound.

#### *FMOD\_DSP\_TYPE\_VSTPLUGIN*

This unit allows the use of Steinberg VST plugins

#### *FMOD\_DSP\_TYPE\_WINAMPPLUGIN*

This unit allows the use of Nullsoft Winamp plugins

#### *FMOD\_DSP\_TYPE\_ITECHO*

This unit produces an echo on the sound and fades out at the desired rate as is used in Impulse Tracker.

#### *FMOD\_DSP\_TYPE\_COMPRESSOR*

This unit implements dynamic compression (linked multichannel, wideband)

#### *FMOD\_DSP\_TYPE\_SFXREVERB*

This unit implements SFX reverb

## *FMOD\_DSP\_TYPE\_LOWPASS\_SIMPLE*

This unit filters sound using a simple lowpass with no resonance, but has flexible cutoff and is fast.

### **Remarks**

To get them to be active, first create the unit, then add it somewhere into the DSP network, either at the front of the network near the soundcard unit to affect the global output (by using `System::getDSPHead`), or on a single channel (using `Channel::getDSPHead`).

### **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

### **See Also**

- [\\_System::createDSPByType](#)

# FMOD\_OPENSTATE

These values describe what state a sound is in after [FMOD\\_NONBLOCKING](#) has been used to open it.?

## Enumeration

```
typedef enum {
    FMOD_OPENSTATE_READY,
    FMOD_OPENSTATE_LOADING,
    FMOD_OPENSTATE_ERROR,
    FMOD_OPENSTATE_CONNECTING,
    FMOD_OPENSTATE_BUFFERING,
    FMOD_OPENSTATE_SEEKING,
    FMOD_OPENSTATE_MAX
} FMOD_OPENSTATE;
```

## Values

*FMOD\_OPENSTATE\_READY*

Opened and ready to play.

*FMOD\_OPENSTATE\_LOADING*

Initial load in progress.

*FMOD\_OPENSTATE\_ERROR*

Failed to open - file not found, out of memory etc. See return value of [Sound::getOpenState](#) for what happened.

*FMOD\_OPENSTATE\_CONNECTING*

Connecting to remote host (internet sounds only).

*FMOD\_OPENSTATE\_BUFFERING*

Buffering data.

*FMOD\_OPENSTATE\_SEEKING*

Seeking to subsound and re-flushing stream buffer.

*FMOD\_OPENSTATE\_MAX*

Maximum number of open state types.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

**See Also**

- [\\_Sound::getOpenState](#)
- [FMOD\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_OUTPUTTYPE

These output types are used with [System::setOutput/System::getOutput](#), to choose which output method to use.?

## Enumeration

```
typedef enum {
    FMOD_OUTPUTTYPE_AUTODETECT,
    FMOD_OUTPUTTYPE_UNKNOWN,
    FMOD_OUTPUTTYPE_NOSOUND,
    FMOD_OUTPUTTYPE_WAVWRITER,
    FMOD_OUTPUTTYPE_NOSOUND_NRT,
    FMOD_OUTPUTTYPE_WAVWRITER_NRT,
    FMOD_OUTPUTTYPE_SOFTWARE,
    FMOD_OUTPUTTYPE_WIIM,
    FMOD_OUTPUTTYPE_OREAL,
    FMOD_OUTPUTTYPE_WASAP,
    FMOD_OUTPUTTYPE_ASIO,
    FMOD_OUTPUTTYPE_OSS,
    FMOD_OUTPUTTYPE_ALSA,
    FMOD_OUTPUTTYPE_ESD,
    FMOD_OUTPUTTYPE_SOUNDMAAGER,
    FMOD_OUTPUTTYPE_COREAUDIO,
    FMOD_OUTPUTTYPE_XBOX,
    FMOD_OUTPUTTYPE_B2,
    FMOD_OUTPUTTYPE_B3,
    FMOD_OUTPUTTYPE_GC,
    FMOD_OUTPUTTYPE_XBOX360,
    FMOD_OUTPUTTYPE_BP,
    FMOD_OUTPUTTYPE_WII,
    FMOD_OUTPUTTYPE_MAX
} FMOD_OUTPUTTYPE;
```

## Values

*FMOD\_OUTPUTTYPE\_AUTODETECT*

Picks the best output mode for the platform. This is the default.

*FMOD\_OUTPUTTYPE\_UNKNOWN*

All - 3rd party plugin, unknown. This is for use with [System::getOutput](#) only.

*FMOD\_OUTPUTTYPE\_NOSOUND*

All - All calls in this mode succeed but make no sound.

*FMOD\_OUTPUTTYPE\_WAVWRITER*

All - Writes output to fmodoutput.wav by default. Use the 'extradriverdata' parameter in [System::init](#), by simply passing the filename as a string, to set the wav filename.

*FMOD\_OUTPUTTYPE\_NOSOUND\_NRT*

All - Non-realtime version of [FMOD\\_OUTPUTTYPE\\_NOSOUND](#). User can drive mixer with [System:update](#) at whatever rate they want.

*FMOD\_OUTPUTTYPE\_WAVWRITER\_NRT*

All - Non-realtime version of [FMOD\\_OUTPUTTYPE\\_WAVWRITER](#). User can drive mixer with [System:update](#) at whatever rate they want.

*FMOD\_OUTPUTTYPE\_DSOUND*

Win32/Win64 - DirectSound output. Use this to get hardware accelerated 3d audio and EAX Reverb support. (Default on Windows)

*FMOD\_OUTPUTTYPE\_WINMM*

Win32/Win64 - Windows Multimedia output.

*FMOD\_OUTPUTTYPE\_OPENAL*

Win32/Win64 - OpenAL 1.1 output. Use this for lower CPU overhead than [FMOD\\_OUTPUTTYPE\\_DSOUND](#), and also Vista H/W support with Creative Labs cards.

*FMOD\_OUTPUTTYPE\_WASAPI*

Win32 - Windows Audio Session API. (Default on Windows Vista)

*FMOD\_OUTPUTTYPE\_ASIO*

Win32 - Low latency ASIO driver.

*FMOD\_OUTPUTTYPE\_OSS*

Linux - Open Sound System output. (Default on Linux)

*FMOD\_OUTPUTTYPE\_ALSA*

Linux - Advanced Linux Sound Architecture output.

*FMOD\_OUTPUTTYPE\_ESD*

Linux - Enlightenment Sound Daemon output.

*FMOD\_OUTPUTTYPE\_SOUNDMANAGER*

Mac - Macintosh SoundManager output. (Default on Mac carbon library)

*FMOD\_OUTPUTTYPE\_COREAUDIO*

Mac - Macintosh CoreAudio output. (Default on Mac OSX library)

*FMOD\_OUTPUTTYPE\_XBOX*

Xbox - Native hardware output. (Default on Xbox)

## *FMOD\_OUTPUTTYPE\_PS2*

PS2 - Native hardware output. (Default on PS2)

## *FMOD\_OUTPUTTYPE\_PS3*

PS3 - Native hardware output. (Default on PS3)

## *FMOD\_OUTPUTTYPE\_GC*

GameCube - Native hardware output. (Default on GameCube)

## *FMOD\_OUTPUTTYPE\_XBOX360*

Xbox 360 - Native hardware output. (Default on Xbox 360)

## *FMOD\_OUTPUTTYPE\_PSP*

PSP - Native hardware output. (Default on PSP)

## *FMOD\_OUTPUTTYPE\_WII*

Wii - Native hardware output. (Default on Wii)

## *FMOD\_OUTPUTTYPE\_MAX*

Maximum number of output types supported.

## **Remarks**

To pass information to the driver when initializing fmod use the extradriverdata parameter in [System::init](#) for the following reasons.

- *FMOD\_OUTPUTTYPE\_WAVWRITER* - extradriverdata is a pointer to a char \* filename that the wav writer will output to.
- *FMOD\_OUTPUTTYPE\_WAVWRITER\_NRT* - extradriverdata is a pointer to a char \* filename that the wav writer will output to.
- *FMOD\_OUTPUTTYPE\_DSOUND* - extradriverdata is a pointer to a HWND so that FMOD can set the focus on the audio for a particular window.
- *FMOD\_OUTPUTTYPE\_GC* - extradriverdata is a pointer to a FMOD\_GC\_INFO struct. This can be found in fmodgc.h.
- *FMOD\_OUTPUTTYPE\_ALSA* - extradriverdata is a pointer to a char \* argument if required by the chosen ALSA driver. Currently these are the only FMOD drivers that take extra information. Other unknown plugins may have different requirements.

Note! If [FMOD\\_OUTPUTTYPE\\_WAVWRITER\\_NRT](#) or [FMOD\\_OUTPUTTYPE\\_NOSOUND\\_NRT](#) are used, and if the [System::update](#) function is being called very quickly (ie for a non realtime decode) it may be being called too quickly for the FMOD streamer thread to respond to. The result will be a skipping/stuttering output in the captured audio.

To remedy this, disable the FMOD Ex streamer thread, and use [FMOD\\_INIT\\_STREAM\\_FROM\\_UPDATE](#) can be used to avoid skipping in the output stream, as it will lock the mixer and the streamer together in the same thread.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System:setOutput](#)
- [System:getOutput](#)
- [System:setSoftwareFormat](#)
- [System:getSoftwareFormat](#)
- [System:init](#)
- [System:update](#)
- [EMOD\\_INITFLAGS](#)



# FMOD\_PLUGINTYPE

These are plugin types defined for use with the [System::getNumPlugins](#), [System::getPluginInfo](#) and [System::unloadPlugin](#) functions.

## Enumeration

```
enum {  
    FMOD_PLUGINTYPE_OUTPUT,  
    FMOD_PLUGINTYPE_CODEC,  
    FMOD_PLUGINTYPE_DSP,  
    FMOD_PLUGINTYPE_MAX  
};
```

## Values

*FMOD\_PLUGINTYPE\_OUTPUT*

The plugin type is an output module. FMOD mixed audio will play through one of these devices

*FMOD\_PLUGINTYPE\_CODEC*

The plugin type is a file format codec. FMOD will use these codecs to load file formats for playback.

*FMOD\_PLUGINTYPE\_DSP*

The plugin type is a DSP unit. FMOD will use these plugins as part of its DSP network to apply effects to output or generate sound in realtime.

*FMOD\_PLUGINTYPE\_MAX*

Maximum number of plugin types supported.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::getNumPlugins](#)
- [System::getPluginInfo](#)
- [System::unloadPlugin](#)



# FMOD\_RESULT

error codes. Returned from every function.?

## Enumeration

y p d fe nm {  
 MO DOK ,  
 MO DE RRA LEA D DCKE D,  
 MO DE RR A COMMA ND,  
 MO DE RRC DA DE ER ,  
 MO DE RRC DA I NT,  
 MO DE RRC DA I NVA L D E YCE ,  
 MO DE RRC DA NAU DO ,  
 MO DE RRC DA N E YCES ,  
 MO DE RRC DA N ISC ,  
 MO DE RRC DA EA D,  
 MO DE RRC A NE LA LDC ,  
 MO DE RRC A NE LS D E N,  
 MO DE RRCOM ,  
 MO DE RR MA ,  
 MO DE RR S PCO NEC TO N,  
 MO DE RR S P D MAT,  
 MO DE RR S P N TOUN D,  
 MO DE RR S P R N N ,  
 MO DE RR S P DOMA MCO NEC TO S ,  
 MO DE RR F E A D,  
 MO DE RR F E COULD N BE EK ,  
 MO DE RR F E DSKE EC E D,  
 MO DE RR F E EO F,  
 MO DE RR F E N TOUN D,  
 MO DE RR F E U MA NE D,  
 MO DE RR D MAT,  
 MO DE RR HTTP,  
 MO DE RR HTTPACCESS ,  
 MO DE RR HTTP PR X AUTH,  
 MO DE RR HTTPSE RE RE R R,  
 MO DE RR HTTP TMEOUT,  
 MO DE RRI N TAIL A TO N,  
 MO DE RRI N TAIL E D,  
 MO DE RRI NE RM L,  
 MO DE RRI NV L DA DDESS ,  
 MO DE RRI NV L D FDA T,  
 MO DE RRI NV L D A NDE ,  
 MO DE RRI NV L D A AM ,  
 MO DE RRI NV L DS EAKE R,  
 MO DE RRI NV L D EC D R,  
 MO DE RRI RX,  
 MO DE RRMA A U D BE ,  
 MO DE RRMEMO E ,  
 MO DE RRMEMO E IO P,  
 MO DE RRMEMO E S AM ,  
 MO DE RRMEMO E CA NTDI NT,  
 MO DE RR NE S 2D,  
 MO DE RR NE S 3 D,  
 MO DE RR NE S A RDA R ,  
 MO DE RR NE SSO FWA R ,  
 MO DE RR N TCO NEC T,

```

MO DE RR N TSOCKE TE RD R,
MO DE RR N TURL,
MO DE RR N TWOULD BDCK ,
MO DE RR N TEA N ,
MO DE RROUTU TA LDCA E D,
MO DE RROUTU TC EA E BFE R,
MO DE RROUTU T DE E KA LL,
MO DE RROUTU TE NME A TO N,
MO DE RROUTU T D MA T,
MO DE RROUTU TI N T,
MO DE RROUTU T N A RWA E ,
MO DE RROUTU T NSO FWA E ,
MO DE RR A N,
MO DE RR PLGI N,
MO DE RR PLGI NMISSI N ,
MO DE RR PLGI N ESOU RE ,
MO DE RR PLGI NI S A NES ,
MO DE RR ECO RD,
MO DE RR E E RBI S A NE ,
MO DE RRSUBOUNS ,
MO DE RRSUBOUNDA LDCA E D,
MO DE RR AG N TOUN D,
MO DE RR DOMA MC A N E S ,
MO DE RR UNMPEME NE D,
MO DE RR UNNTA L E D,
MO DE RR UNSUPD RE D,
MO DE RRUPA E ,
MO DE RR E RIO N,
MO DE RRE E NT AI E D,
MO DE RRE E NTI NE RA L,
MO DE RRE E NTI NDO NZ ,
MO DE RRE E NTMA S TEAMS ,
MO DE RRE E NTMISMA T H,
MO DE RRE E NT AMECO NFIC T,
MO DE RRE E NT N TOUN D
} MO D ESU LT

```

## Values

*FMOD\_OK*

No errors.

*FMOD\_ERR\_ALREADYLOCKED*

Tried to call lock a second time before unlock was called.

*FMOD\_ERR\_BADCOMMAND*

Tried to call a function on a data type that does not allow this type of functionality (ie calling Sound::lock on a streaming sound).

*FMOD\_ERR\_CDDA\_DRIVERS*

Neither NTSCSI nor ASPI could be initialised.

*FMOD\_ERR\_CDDA\_INIT*

An error occurred while initialising the CDDA subsystem.

*FMOD\_ERR\_CDDA\_INVALID\_DEVICE*

Couldn't find the specified device.

*FMOD\_ERR\_CDDA\_NOAUDIO*

No audio tracks on the specified disc.

*FMOD\_ERR\_CDDA\_NODEVICES*

No CD/DVD devices were found.

*FMOD\_ERR\_CDDA\_NODISC*

No disc present in the specified drive.

*FMOD\_ERR\_CDDA\_READ*

A CDDA read error occurred.

*FMOD\_ERR\_CHANNEL\_ALLOC*

Error trying to allocate a channel.

*FMOD\_ERR\_CHANNEL\_STOLEN*

The specified channel has been reused to play another sound.

*FMOD\_ERR\_COM*

A Win32 COM related error occurred. COM failed to initialize or a QueryInterface failed meaning a Windows codec or driver was not installed properly.

*FMOD\_ERR\_DMA*

DMA Failure. See debug output for more information.

*FMOD\_ERR\_DSP\_CONNECTION*

DSP connection error. Connection possibly caused a cyclic dependency.

*FMOD\_ERR\_DSP\_FORMAT*

DSP Format error. A DSP unit may have attempted to connect to this network with the wrong format.

*FMOD\_ERR\_DSP\_NOTFOUND*

DSP connection error. Couldn't find the DSP unit specified.

*FMOD\_ERR\_DSP\_RUNNING*

DSP error. Cannot perform this operation while the network is in the middle of running. This will most likely happen if a connection or disconnection is attempted in a DSP callback.

### *FMOD\_ERR\_DSP\_TOOMANYCONNECTIONS*

DSP connection error. The unit being connected to or disconnected should only have 1 input or output.

### *FMOD\_ERR\_FILE\_BAD*

Error loading file.

### *FMOD\_ERR\_FILE\_COULDNOTSEEK*

Couldn't perform seek operation. This is a limitation of the medium (ie netstreams) or the file format.

### *FMOD\_ERR\_FILE\_DISKEJECTED*

Media was ejected while reading.

### *FMOD\_ERR\_FILE\_EOF*

End of file unexpectedly reached while trying to read essential data (truncated data?).

### *FMOD\_ERR\_FILE\_NOTFOUND*

File not found.

### *FMOD\_ERR\_FILE\_UNWANTED*

Unwanted file access occurred.

### *FMOD\_ERR\_FORMAT*

Unsupported file or audio format.

### *FMOD\_ERR\_HTTP*

A HTTP error occurred. This is a catch-all for HTTP errors not listed elsewhere.

### *FMOD\_ERR\_HTTP\_ACCESS*

The specified resource requires authentication or is forbidden.

### *FMOD\_ERR\_HTTP\_PROXY\_AUTH*

Proxy authentication is required to access the specified resource.

### *FMOD\_ERR\_HTTP\_SERVER\_ERROR*

A HTTP server error occurred.

### *FMOD\_ERR\_HTTP\_TIMEOUT*

The HTTP request timed out.

### *FMOD\_ERR\_INITIALIZATION*

FMOD was not initialized correctly to support this function.

### *FMOD\_ERR\_INITIALIZED*

Cannot call this command after System::init.

### *FMOD\_ERR\_INTERNAL*

An error occurred that wasn't supposed to. Contact support.

### *FMOD\_ERR\_INVALID\_ADDRESS*

On Xbox 360, this memory address passed to FMOD must be physical, (ie allocated with XPhysicalAlloc.)

### *FMOD\_ERR\_INVALID\_FLOAT*

Value passed in was a NaN, Inf or denormalized float.

### *FMOD\_ERR\_INVALID\_HANDLE*

An invalid object handle was used.

### *FMOD\_ERR\_INVALID\_PARAM*

An invalid parameter was passed to this function.

### *FMOD\_ERR\_INVALID\_SPEAKER*

An invalid speaker was passed to this function based on the current speaker mode.

### *FMOD\_ERR\_INVALID\_VECTOR*

The vectors passed in are not unit length, or perpendicular.

### *FMOD\_ERR\_IRX*

PS2 only. fmindex.irx failed to initialize. This is most likely because you forgot to load it.

### *FMOD\_ERR\_MAXAUDIBLE*

Reached maximum audible playback count for this sound's soundgroup.

### *FMOD\_ERR\_MEMORY*

Not enough memory or resources.

### *FMOD\_ERR\_MEMORY\_IOP*

PS2 only. Not enough memory or resources on PlayStation 2 IOP ram.

### *FMOD\_ERR\_MEMORY\_SRAM*

Not enough memory or resources on console sound ram.

### *FMOD\_ERR\_MEMORY\_CANTPOINT*

Can't use [FMOD\\_OPENMEMORY\\_POINT](#) on non PCM source data, or non mp3/xma/adpcm data if [FMOD\\_CREATECOMPRESSED\\_SAMPLE](#) was used.

#### *FMOD\_ERR\_NEEDS2D*

Tried to call a command on a 3d sound when the command was meant for 2d sound.

#### *FMOD\_ERR\_NEEDS3D*

Tried to call a command on a 2d sound when the command was meant for 3d sound.

#### *FMOD\_ERR\_NEEDSHARDWARE*

Tried to use a feature that requires hardware support. (ie trying to play a VAG compressed sound in software on PS2).

#### *FMOD\_ERR\_NEEDSSOFTWARE*

Tried to use a feature that requires the software engine. Software engine has either been turned off, or command was executed on a hardware channel which does not support this feature.

#### *FMOD\_ERR\_NET\_CONNECT*

Couldn't connect to the specified host.

#### *FMOD\_ERR\_NET\_SOCKET\_ERROR*

A socket error occurred. This is a catch-all for socket-related errors not listed elsewhere.

#### *FMOD\_ERR\_NET\_URL*

The specified URL couldn't be resolved.

#### *FMOD\_ERR\_NET\_WOULD\_BLOCK*

Operation on a non-blocking socket could not complete immediately.

#### *FMOD\_ERR\_NOTREADY*

Operation could not be performed because specified sound is not ready.

#### *FMOD\_ERR\_OUTPUT\_ALLOCATED*

Error initializing output device, but more specifically, the output device is already in use and cannot be reused.

#### *FMOD\_ERR\_OUTPUT\_CREATEBUFFER*

Error creating hardware sound buffer.

#### *FMOD\_ERR\_OUTPUT\_DRIVERCALL*

A call to a standard soundcard driver failed, which could possibly mean a bug in the driver or resources were missing or exhausted.

#### *FMOD\_ERR\_OUTPUT\_ENUMERATION*



Error enumerating the available driver list. List may be inconsistent due to a recent device addition or removal.

#### *FMOD\_ERR\_OUTPUT\_FORMAT*

Soundcard does not support the minimum features needed for this soundsystem (16bit stereo output).

#### *FMOD\_ERR\_OUTPUT\_INIT*

Error initializing output device.

#### *FMOD\_ERR\_OUTPUT\_NOHARDWARE*

FMOD\_HARDWARE was specified but the sound card does not have the resources necessary to play it.

#### *FMOD\_ERR\_OUTPUT\_NOSOFTWARE*

Attempted to create a software sound but no software channels were specified in System::init.

#### *FMOD\_ERR\_PAN*

Panning only works with mono or stereo sound sources.

#### *FMOD\_ERR\_PLUGIN*

An unspecified error has been returned from a 3rd party plugin.

#### *FMOD\_ERR\_PLUGIN\_MISSING*

A requested output, dsp unit type or codec was not available.

#### *FMOD\_ERR\_PLUGIN\_RESOURCE*

A resource that the plugin requires cannot be found. (ie the DLS file for MIDI playback)

#### *FMOD\_ERR\_PLUGIN\_INSTANCES*

The number of allowed instances of a plugin has been exceeded.

#### *FMOD\_ERR\_RECORD*

An error occurred trying to initialize the recording device.

#### *FMOD\_ERR\_REVERB\_INSTANCE*

Specified Instance in FMOD\_REVERB\_PROPERTIES couldn't be set. Most likely because another application has locked the EAX4 FX slot.

#### *FMOD\_ERR\_SUBSOUNDS*

The error occurred because the sound referenced contains subsounds. (ie you cannot play the parent sound as a static sample, only its subsounds.)

#### *FMOD\_ERR\_SUBSOUND\_ALLOCATED*

This subsound is already being used by another sound, you cannot have more than one parent to a sound. Null out the other parent's entry first.

#### *FMOD\_ERR\_TAGNOTFOUND*

The specified tag could not be found or there are no tags.

#### *FMOD\_ERR\_TOOMANYCHANNELS*

The sound created exceeds the allowable input channel count. This can be increased using the `maxinputchannels` parameter in `System::setSoftwareFormat`.

#### *FMOD\_ERR\_UNIMPLEMENTED*

Something in FMOD hasn't been implemented when it should be! contact support!

#### *FMOD\_ERR\_UNINITIALIZED*

This command failed because `System::init` or `System::setDriver` was not called.

#### *FMOD\_ERR\_UNSUPPORTED*

A command issued was not supported by this object. Possibly a plugin without certain callbacks specified.

#### *FMOD\_ERR\_UPDATE*

An error caused by `System::update` occurred.

#### *FMOD\_ERR\_VERSION*

The version number of this file format is not supported.

#### *FMOD\_ERR\_EVENT\_FAILED*

An Event failed to be retrieved, most likely due to 'just fail' being specified as the max playbacks behavior.

#### *FMOD\_ERR\_EVENT\_INTERNAL*

An error occurred that wasn't supposed to. See debug log for reason.

#### *FMOD\_ERR\_EVENT\_INFOONLY*

Can't execute this command on an `EVENT_INFOONLY` event.

#### *FMOD\_ERR\_EVENT\_MAXSTREAMS*

Event failed because 'Max streams' was hit when `FMOD_INIT_FAIL_ON_MAXSTREAMS` was specified.

#### *FMOD\_ERR\_EVENT\_MISMATCH*

FSB mismatches the FEV it was compiled with or FEV was built for a different platform.

#### *FMOD\_ERR\_EVENT\_NAMECONFLICT*

A category with the same name already exists.

*FMOD\_ERR\_EVENT\_NOTFOUND*

The requested event, event group, event category or event property could not be found.

### **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_SOUNDGROUP\_BEHAVIOR

These flags are used with [SoundGroup::setMaxAudibleBehavior](#) to determine what happens when more sounds are played than are specified with [SoundGroup::setMaxAudible](#).

## Enumeration

```
typedef enum {
    FMOD_SOUNDGROUP_BEHAVIOR_FAIL,
    FMOD_SOUNDGROUP_BEHAVIOR_MUTE,
    FMOD_SOUNDGROUP_BEHAVIOR_STEALLOWEST,
    FMOD_SOUNDGROUP_BEHAVIOR_MAX
}
```

## Values

*FMOD\_SOUNDGROUP\_BEHAVIOR\_FAIL*

Any sound played that puts the sound count over the [SoundGroup::setMaxAudible](#) setting, will simply fail during System::playSound.

*FMOD\_SOUNDGROUP\_BEHAVIOR\_MUTE*

Any sound played that puts the sound count over the [SoundGroup::setMaxAudible](#) setting, will be silent, then if another sound in the group stops the sound that was silent before becomes audible again.

*FMOD\_SOUNDGROUP\_BEHAVIOR\_STEALLOWEST*

Any sound played that puts the sound count over the [SoundGroup::setMaxAudible](#) setting, will steal the quietest / least important sound playing in the group.

*FMOD\_SOUNDGROUP\_BEHAVIOR\_MAX*

Maximum number of open state types.

## Remarks

When using [FMOD\\_SOUNDGROUP\\_BEHAVIOR\\_MUTE](#), [SoundGroup::setMuteFadeSpeed](#) can be used to stop a sudden transition. Instead, the time specified will be used to cross fade between the sounds that go silent and the ones that become audible.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable,

PlayStation 3, Wii, Solaris

#### See Also

- [\\_SoundGroup::setMaxAudibleBehavior](#)
- [\\_SoundGroup::getMaxAudibleBehavior](#)
- [\\_SoundGroup::setMaxAudible](#)
- [\\_SoundGroup::getMaxAudible](#)
- [\\_SoundGroup::setMuteFadeSpeed](#)
- [\\_SoundGroup::getMuteFadeSpeed](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_SOUND\_FORMAT

These definitions describe the native format of the hardware or software buffer that will be used.?

## Enumeration

```
typedef enum {
    FMOD_SOUND_FORMAT_NONE,
    FMOD_SOUND_FORMAT_PCM8,
    FMOD_SOUND_FORMAT_PCM16,
    FMOD_SOUND_FORMAT_PCM24,
    FMOD_SOUND_FORMAT_PCM32,
    FMOD_SOUND_FORMAT_PCMFLOAT,
    FMOD_SOUND_FORMAT_GCADPCM,
    FMOD_SOUND_FORMAT_TMAA_DPCM,
    FMOD_SOUND_FORMAT_VAG,
    FMOD_SOUND_FORMAT_XMA,
    FMOD_SOUND_FORMAT_XMPEG,
    FMOD_SOUND_FORMAT_TMA_X
}
```

## Values

*FMOD\_SOUND\_FORMAT\_NONE*

Unitialized / unknown.

*FMOD\_SOUND\_FORMAT\_PCM8*

8bit integer PCM data.

*FMOD\_SOUND\_FORMAT\_PCM16*

16bit integer PCM data.

*FMOD\_SOUND\_FORMAT\_PCM24*

24bit integer PCM data.

*FMOD\_SOUND\_FORMAT\_PCM32*

32bit integer PCM data.

*FMOD\_SOUND\_FORMAT\_PCMFLOAT*

32bit floating point PCM data.

*FMOD\_SOUND\_FORMAT\_GCADPCM*

Compressed GameCube DSP data.

*FMOD\_SOUND\_FORMAT\_IMAADPCM*

Compressed IMA ADPCM / Xbox ADPCM data.

*FMOD\_SOUND\_FORMAT\_VAG*

Compressed PlayStation 2 / PlayStation Portable ADPCM data.

*FMOD\_SOUND\_FORMAT\_XMA*

Compressed Xbox360 data.

*FMOD\_SOUND\_FORMAT\_MPEG*

Compressed MPEG layer 2 or 3 data.

*FMOD\_SOUND\_FORMAT\_MAX*

Maximum number of sound formats supported.

## **Remarks**

This is the format the native hardware or software buffer will be or is created in.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_System::createSound](#)
- [\\_Sound::getFormat](#)

# FMOD\_SOUND\_TYPE

These definitions describe the type of song being played.?

## Enumeration

```
typedef enum {
    FMOD_SOUND_TYPE_UNKNOWN,
    FMOD_SOUND_TYPE_AAC,
    FMOD_SOUND_TYPE_AIFF,
    FMOD_SOUND_TYPE_ASF,
    FMOD_SOUND_TYPE_AT3,
    FMOD_SOUND_TYPE_CDA,
    FMOD_SOUND_TYPE_DS,
    FMOD_SOUND_TYPE_FAC,
    FMOD_SOUND_TYPE_EB,
    FMOD_SOUND_TYPE_GCDM,
    FMOD_SOUND_TYPE_IT,
    FMOD_SOUND_TYPE_MID,
    FMOD_SOUND_TYPE_MOD,
    FMOD_SOUND_TYPE_MEG,
    FMOD_SOUND_TYPE_OGG_VRES,
    FMOD_SOUND_TYPE_PLAYLIST,
    FMOD_SOUND_TYPE_RAW,
    FMOD_SOUND_TYPE_S3M,
    FMOD_SOUND_TYPE_SF2,
    FMOD_SOUND_TYPE_USER,
    FMOD_SOUND_TYPE_WAV,
    FMOD_SOUND_TYPE_XM,
    FMOD_SOUND_TYPE_XMA,
    FMOD_SOUND_TYPE_XG,
    FMOD_SOUND_TYPE_MAX
} FMOD_SOUND_TYPE;
```

## Values

*FMOD\_SOUND\_TYPE\_UNKNOWN*

3rd party / unknown plugin format.

*FMOD\_SOUND\_TYPE\_AAC*

AAC. Currently unsupported.

*FMOD\_SOUND\_TYPE\_AIFF*

AIFF.

*FMOD\_SOUND\_TYPE\_ASF*

Microsoft Advanced Systems Format (ie WMA/ASF/WMV).

*FMOD\_SOUND\_TYPE\_AT3*



Sony ATRAC 3 format

*FMOD\_SOUND\_TYPE\_CDDA*

Digital CD audio.

*FMOD\_SOUND\_TYPE\_DLS*

Sound font / downloadable sound bank.

*FMOD\_SOUND\_TYPE\_FLAC*

FLAC lossless codec.

*FMOD\_SOUND\_TYPE\_FSB*

FMOD Sample Bank.

*FMOD\_SOUND\_TYPE\_GCADPCM*

GameCube ADPCM

*FMOD\_SOUND\_TYPE\_IT*

Impulse Tracker.

*FMOD\_SOUND\_TYPE\_MIDI*

MIDI.

*FMOD\_SOUND\_TYPE\_MOD*

Protracker / Fasttracker MOD.

*FMOD\_SOUND\_TYPE\_MPEG*

MP2/MP3 MPEG.

*FMOD\_SOUND\_TYPE\_OGGVORBIS*

Ogg vorbis.

*FMOD\_SOUND\_TYPE\_PLAYLIST*

Information only from ASX/PLS/M3U/WAX playlists

*FMOD\_SOUND\_TYPE\_RAW*

Raw PCM data.

*FMOD\_SOUND\_TYPE\_S3M*

ScreamTracker 3.

*FMOD\_SOUND\_TYPE\_SF2*

Sound font 2 format.

*FMOD\_SOUND\_TYPE\_USER*

User created sound.

*FMOD\_SOUND\_TYPE\_WAV*

Microsoft WAV.

*FMOD\_SOUND\_TYPE\_XM*

FastTracker 2 XM.

*FMOD\_SOUND\_TYPE\_XMA*

Xbox360 XMA

*FMOD\_SOUND\_TYPE\_VAG*

PlayStation 2 / PlayStation Portable adpcm VAG format.

*FMOD\_SOUND\_TYPE\_MAX*

Maximum number of sound types supported.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_Sound::getFormat](#)

# FMOD\_SPEAKER

These are speaker types defined for use with the [Channel::setSpeakerLevels](#) command.?It can also be used for speaker placement in the [System::set3DSpeakerPosition](#) command.?

## Enumeration

```
typedef enum {
    FMOD_SPEAKER_FRONT_LEFT,
    FMOD_SPEAKER_FRONT_RIGHT,
    FMOD_SPEAKER_FRONT_CENTER,
    FMOD_SPEAKER_LOW_FREQUENCY,
    FMOD_SPEAKER_BACK_LEFT,
    FMOD_SPEAKER_BACK_RIGHT,
    FMOD_SPEAKER_SIDE_LEFT,
    FMOD_SPEAKER_SIDE_RIGHT,
    FMOD_SPEAKER_MAX,
    FMOD_SPEAKER_NULL,
    FMOD_SPEAKER_RESERVED,
    FMOD_SPEAKER_COUNT
}
```

## Values

*FMOD\_SPEAKER\_FRONT\_LEFT*

*FMOD\_SPEAKER\_FRONT\_RIGHT*

*FMOD\_SPEAKER\_FRONT\_CENTER*

*FMOD\_SPEAKER\_LOW\_FREQUENCY*

*FMOD\_SPEAKER\_BACK\_LEFT*

*FMOD\_SPEAKER\_BACK\_RIGHT*

*FMOD\_SPEAKER\_SIDE\_LEFT*

*FMOD\_SPEAKER\_SIDE\_RIGHT*

*FMOD\_SPEAKER\_MAX*

Maximum number of speaker types supported.

## *FMOD\_SPEAKER\_MONO*

For use with [FMOD\\_SPEAKERMODE\\_MONO](#) and `Channel::SetSpeakerLevels`. Mapped to same value as [FMOD\\_SPEAKER\\_FRONT\\_LEFT](#).

## *FMOD\_SPEAKER\_NULL*

A non speaker. Use this to send.

## *FMOD\_SPEAKER\_SBL*

For use with [FMOD\\_SPEAKERMODE\\_7POINT1](#) on PS3 where the extra speakers are surround back inside of side speakers.

## *FMOD\_SPEAKER\_SBR*

For use with [FMOD\\_SPEAKERMODE\\_7POINT1](#) on PS3 where the extra speakers are surround back inside of side speakers.

## **Remarks**

If you are using [FMOD\\_SPEAKERMODE\\_RAW](#) and speaker assignments are meaningless, just cast a raw integer value to this type.

For example ([FMOD\\_SPEAKER](#))7 would use the 7th speaker (also the same as [FMOD\\_SPEAKER\\_SIDE\\_RIGHT](#)).

Values higher than this can be used if an output system has more than 8 speaker types / output channels. 15 is the current maximum.

NOTE: On Playstation 3 in 7.1, the extra 2 speakers are not side left/side right, they are 'surround back left'/'surround back right' which locate the speakers behind the listener instead of to the sides like on PC. [FMOD\\_SPEAKER\\_SBL](#)/[FMOD\\_SPEAKER\\_SBR](#) are provided to make it clearer what speaker is being addressed on that platform.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [FMOD\\_SPEAKERMODE](#)
- [Channel::setSpeakerLevels](#)
- [Channel::getSpeakerLevels](#)
- [System::set3DSpeakerPosition](#)
- [System::get3DSpeakerPosition](#)

# FMOD\_SPEAKERMATYPE

When creating a multichannel sound, FMOD will pan them to their default speaker locations, for example a 6 channel sound will default to one channel per 5.1 output speaker.

?Another example is a stereo sound. It will default to left = front left, right = front right.

?

?This is for sounds that are not 'default'. For example you might have a sound that is 6 channels but actually made up of 3 stereo pairs, that should all be located in front left, front right only.?

## Enumeration

```
typedef enum {
    FMOD_SPEAKERMATYPE_DEFAULT,
    FMOD_SPEAKERMATYPE_ALLMONO,
    FMOD_SPEAKERMATYPE_ALLSTEREO
} FMOD_SPEAKERMATYPE;
```

## Values

*FMOD\_SPEAKERMATYPE\_DEFAULT*

This is the default, and just means FMOD decides which speakers it puts the source channels.

*FMOD\_SPEAKERMATYPE\_ALLMONO*

This means the sound is made up of all mono sounds. All voices will be panned to the front center by default in this case.

*FMOD\_SPEAKERMATYPE\_ALLSTEREO*

This means the sound is made up of all stereo sounds. All voices will be panned to front left and front right alternating every second channel.

## Remarks

For full flexibility of speaker assignments, use [Channel::setSpeakerLevels](#). This functionality is cheaper, uses less memory and easier to use.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [FMOD\\_CREATEINDEXINFO](#)

- [\\_Channel::setSpeakerLevels](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_SPEAKERMODE

These are speaker types defined for use with the [System::setSpeakerMode](#) or [System::getSpeakerMode](#) command.?

## Enumeration

```
typedef enum {  
    FMOD_SPEAKERMODE_RAW,  
    FMOD_SPEAKERMODE_MONO,  
    FMOD_SPEAKERMODE_STEREO,  
    FMOD_SPEAKERMODE_QUAD,  
    FMOD_SPEAKERMODE_SURROUND,  
    FMOD_SPEAKERMODE_5POINT1,  
    FMOD_SPEAKERMODE_7POINT1,  
    FMOD_SPEAKERMODE_PROLOGIC,  
    FMOD_SPEAKERMODE_MAX  
}
```

## Values

*FMOD\_SPEAKERMODE\_RAW*

There is no specific speakermode. Sound channels are mapped in order of input to output. Use [System::setSoftwareFormat](#) to specify speaker count. See remarks for more information.

*FMOD\_SPEAKERMODE\_MONO*

The speakers are monaural.

*FMOD\_SPEAKERMODE\_STEREO*

The speakers are stereo (DEFAULT).

*FMOD\_SPEAKERMODE\_QUAD*

4 speaker setup. This includes front left, front right, rear left, rear right.

*FMOD\_SPEAKERMODE\_SURROUND*

5 speaker setup. This includes front left, front right, center, rear left, rear right.

*FMOD\_SPEAKERMODE\_5POINT1*

5.1 speaker setup. This includes front left, front right, center, rear left, rear right and a subwoofer.

*FMOD\_SPEAKERMODE\_7POINT1*

7.1 speaker setup. This includes front left, front right, center, rear left, rear right, side left, side right and a subwoofer.

*FMOD\_SPEAKERMODE\_PROLOGIC*

Stereo output, but data is encoded in a way that is picked up by a Prologic/Prologic2 decoder and split into a 5.1 speaker setup.

## *FMOD\_SPEAKERMODE\_MAX*

Maximum number of speaker modes supported.

## **Remarks**

These are important notes on speaker modes in regards to sounds created with [FMOD\\_SOFTWARE](#).

Note below the phrase 'sound channels' is used. These are the subchannels inside a sound, they are not related and have nothing to do with the FMOD class "Channel".

For example a mono sound has 1 sound channel, a stereo sound has 2 sound channels, and an AC3 or 6 channel wav file have 6 "sound channels".

## *FMOD\_SPEAKERMODE\_RAW*

-----

This mode is for output devices that are not specifically mono/stereo/quad/surround/5.1 or 7.1, but are multichannel. Use [System::setSoftwareFormat](#) to specify the number of speakers you want to address, otherwise it will default to 2 (stereo).

Sound channels map to speakers sequentially, so a mono sound maps to output speaker 0, stereo sound maps to output speaker 0 ?

The user assumes knowledge of the speaker order. [FMOD\\_SPEAKER](#) enumerations may not apply, so raw channel indices should be used.

Multichannel sounds map input channels to output channels 1:1.

Channel::setPan and Channel::setSpeakerMix do not work.

Speaker levels must be manually set with [Channel::setSpeakerLevels](#).

## *FMOD\_SPEAKERMODE\_MONO*

-----

This mode is for a 1 speaker arrangement.

Panning does not work in this speaker mode.

Mono, stereo and multichannel sounds have each sound channel played on the one speaker unity.

Mix behavior for multichannel sounds can be set with [Channel::setSpeakerLevels](#).

Channel::setSpeakerMix does not work.

## *FMOD\_SPEAKERMODE\_STEREO*

-----

This mode is for 2 speaker arrangements that have a left and right speaker.

- Mono sounds default to an even distribution between left and right. They can be panned with Channel::setPan.
- Stereo sounds default to the middle, or full left in the left speaker and full right in the right speaker.
- They can be cross faded with Channel::setPan.
- Multichannel sounds have each sound channel played on each speaker at unity.
- Mix behavior for multichannel sounds can be set with [Channel::setSpeakerLevels](#).
- Channel::setSpeakerMix works but only front left and right parameters are used, the rest are ignored.

## *FMOD\_SPEAKERMODE\_QUAD*

-----

This mode is for 4 speaker arrangements that have a front left, front right, rear left and a rear right speaker.

- Mono sounds default to an even distribution between front left and front right. They can be panned with Channel::setPan.
- Stereo sounds default to the left sound channel played on the front left, and the right sound channel played on the front right.



- They can be cross faded with `Channel::setPan`.
- Multichannel sounds default to all of their sound channels being played on each speaker in order of input.
- Mix behavior for multichannel sounds can be set with [Channel::setSpeakerLevels](#).
- `Channel::setSpeakerMix` works but side left, side right, center and lfe are ignored.

## FMOD\_SPEAKERMODE\_SURROUND

-----

This mode is for 5 speaker arrangements that have a left/right/center/rear left/rear right.

- Mono sounds default to the center speaker. They can be panned with `Channel::setPan`.
- Stereo sounds default to the left sound channel played on the front left, and the right sound channel played on the front right.
- They can be cross faded with `Channel::setPan`.
- Multichannel sounds default to all of their sound channels being played on each speaker in order of input.
- Mix behavior for multichannel sounds can be set with [Channel::setSpeakerLevels](#).
- `Channel::setSpeakerMix` works but side left / side right are ignored.

## FMOD\_SPEAKERMODE\_5POINT1

-----

This mode is for 5.1 speaker arrangements that have a left/right/center/rear left/rear right and a subwoofer speaker.

- Mono sounds default to the center speaker. They can be panned with `Channel::setPan`.
- Stereo sounds default to the left sound channel played on the front left, and the right sound channel played on the front right.
- They can be cross faded with `Channel::setPan`.
- Multichannel sounds default to all of their sound channels being played on each speaker in order of input.
- Mix behavior for multichannel sounds can be set with [Channel::setSpeakerLevels](#).
- `Channel::setSpeakerMix` works but side left / side right are ignored.

## FMOD\_SPEAKERMODE\_7POINT1

-----

This mode is for 7.1 speaker arrangements that have a left/right/center/rear left/rear right/side left/side right and a subwoofer speaker.

- Mono sounds default to the center speaker. They can be panned with `Channel::setPan`.
- Stereo sounds default to the left sound channel played on the front left, and the right sound channel played on the front right.
- They can be cross faded with `Channel::setPan`.
- Multichannel sounds default to all of their sound channels being played on each speaker in order of input.
- Mix behavior for multichannel sounds can be set with [Channel::setSpeakerLevels](#).
- `Channel::setSpeakerMix` works and every parameter is used to set the balance of a sound in any speaker.

## FMOD\_SPEAKERMODE\_PROLOGIC

-----

This mode is for mono, stereo, 5.1 and 7.1 speaker arrangements, as it is backwards and forwards compatible with stereo, but to get a surround effect a Dolby Prologic or Prologic 2 hardware decoder / amplifier is needed. Pan behavior is the same as [FMOD\\_SPEAKERMODE\\_5POINT1](#).

If this function is called the numoutputchannels setting in [System::setSoftwareFormat](#) is overwritten.

For 3D sounds, panning is determined at runtime by the 3D subsystem based on the speaker mode to determine which speaker the sound should be placed in.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

### See Also

- [System::setSpeakerMode](#)
- [System::getSpeakerMode](#)
- [System::getDriverCaps](#)
- [System::setSoftwareFormat](#)
- [Channel::setSpeakerLevels](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_SYSTEM\_CALLBACKTYPE

These callback types are used with [System::setCallback](#)?

## Enumeration

```
typedef enum {  
    FMOD_SYSTEM_CALLBACKTYPE_DEVICELISTCHANGED,  
    FMOD_SYSTEM_CALLBACKTYPE_MEMORYALLOCATIONFAILED,  
    FMOD_SYSTEM_CALLBACKTYPE_MAX  
}
```

## Values

*FMOD\_SYSTEM\_CALLBACKTYPE\_DEVICELISTCHANGED*

Called from [System::update](#) when the enumerated list of devices has changed.

*FMOD\_SYSTEM\_CALLBACKTYPE\_MEMORYALLOCATIONFAILED*

Called directly when a memory allocation fails somewhere in FMOD.

*FMOD\_SYSTEM\_CALLBACKTYPE\_MAX*

Maximum number of callback types supported.

## Remarks

Each callback has commanddata parameters passed as int unique to the type of callback.

See reference to [FMOD\\_SYSTEM\\_CALLBACK](#) to determine what they might mean for each type of callback.

**Note!** Using [FMOD\\_SYSTEM\\_CALLBACKTYPE\\_MEMORYALLOCATIONFAILED](#) will override any other FMOD::System object registered for this callback.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## See Also

- [System::setCallback](#)
- [FMOD\\_SYSTEM\\_CALLBACK](#)

- [\\_System:update](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_TAGDATATYPE

List of data types that can be returned by Sound::getTag?

## Enumeration

```
typedef enum {
    FMOD_TAGDATATYPE_BINARY,
    FMOD_TAGDATATYPE_INT,
    FMOD_TAGDATATYPE_FLOAT,
    FMOD_TAGDATATYPE_STRING,
    FMOD_TAGDATATYPE_STRING_UTF16,
    FMOD_TAGDATATYPE_STRING_UTF16BE,
    FMOD_TAGDATATYPE_STRING_UTF8,
    FMOD_TAGDATATYPE_CDTOC,
    FMOD_TAGDATATYPE_MAX
} FMOD_TAGDATATYPE;
```

## Values

*FMOD\_TAGDATATYPE\_BINARY*

*FMOD\_TAGDATATYPE\_INT*

*FMOD\_TAGDATATYPE\_FLOAT*

*FMOD\_TAGDATATYPE\_STRING*

*FMOD\_TAGDATATYPE\_STRING\_UTF16*

*FMOD\_TAGDATATYPE\_STRING\_UTF16BE*

*FMOD\_TAGDATATYPE\_STRING\_UTF8*

*FMOD\_TAGDATATYPE\_CDTOC*

*FMOD\_TAGDATATYPE\_MAX*

Maximum number of tag datatypes supported.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

#### See Also

- [\\_Sound::getTag](#)

Version 4.12.03 Built on Feb 18, 2008

# FMOD\_TAGTYPE

List of tag types that could be stored within a sound. These include id3 tags, metadata from netstreams and vorbis/asf data.

## Enumeration

```
typedef enum {
    FMOD_TAGTYPE_UNKNOWN,
    FMOD_TAGTYPE_ID3V1,
    FMOD_TAGTYPE_ID3V2,
    FMOD_TAGTYPE_VORBISCOMMENT,
    FMOD_TAGTYPE_SHOUTCAST,
    FMOD_TAGTYPE_ICECAST,
    FMOD_TAGTYPE_ASF,
    FMOD_TAGTYPE_MIDI,
    FMOD_TAGTYPE_PLAYLIST,
    FMOD_TAGTYPE_FMOD,
    FMOD_TAGTYPE_USER,
    FMOD_TAGTYPE_MAX
} FMOD_TAGTYPE;
```

## Values

*FMOD\_TAGTYPE\_UNKNOWN*

*FMOD\_TAGTYPE\_ID3V1*

*FMOD\_TAGTYPE\_ID3V2*

*FMOD\_TAGTYPE\_VORBISCOMMENT*

*FMOD\_TAGTYPE\_SHOUTCAST*

*FMOD\_TAGTYPE\_ICECAST*

*FMOD\_TAGTYPE\_ASF*

*FMOD\_TAGTYPE\_MIDI*

*FMOD\_TAGTYPE\_PLAYLIST*

*FMOD\_TAGTYPE\_FMOD*

*FMOD\_TAGTYPE\_USER*

*FMOD\_TAGTYPE\_MAX*

Maximum number of tag types supported.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Solaris

## **See Also**

- [\\_Sound::getTag](#)



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# EventSystem Interface

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[EventSystem::setReverbProperties](#)  
[EventSystem::setUserData](#)  
[EventSystem::unload](#)  
[EventSystem::unregisterMemoryFSB](#)  
[EventSystem::update](#)

# EventSystem::createReverb

Creates a 'virtual reverb' object. This object reacts to 3d location and morphs the reverb environment based on how close it is to the reverb object's center.

?Multiple reverb objects can be created to achieve a multi-reverb environment.?

## Syntax

```
FMOD_RESULT fmodSystem::createReverb(  
    EventReverb ** pReverb  
);
```

## Parameters

*reverb*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 3D reverb object is a sphere having 3D attributes (position, minimum distance, maximum distance) and reverb properties.

The properties and 3D attributes of all reverb objects collectively determine, along with the listener's position, the settings of and input gains into a single 3D reverb DSP.

Please note that this only applies to software channels. When the listener is within the sphere of effect of one or more 3d reverbs, the listener's 3D reverb properties are a weighted combination of such 3d reverbs. When the listener is outside all of the reverbs, the 3D reverb setting is set to the default ambient reverb setting.

Use [EventSystem::setReverbAmbientProperties](#) to set a 'background' default reverb environment. This is a reverb that will be morphed to if the listener is not within any virtual reverb zones.

By default the ambient reverb is set to 'off'.

Creating multiple reverb objects does not impact performance. These are 'virtual reverbs'. There will still be only 1 physical reverb DSP running that just morphs between the different virtual reverbs.

[EventSystem::setReverbProperties](#) can still be used in conjunction with the 3d based virtual reverb system. This allows 2d sounds to have reverb. If this call is used at the same time virtual reverb objects are active, 2 physical reverb dsp's will be used, incurring a small memory and cpu hit.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::setReverbAmbientProperties](#)
- [EventSystem::getReverbAmbientProperties](#)
- [EventSystem::setReverbProperties](#)
- [EventSystem::getReverbProperties](#)
- [EventReverb::release](#)

# EventSystem::get3DListenerAttributes

This retrieves the position, velocity and orientation of the specified 3D sound listener.?

## Syntax

```
FMOD_RESULT FModSystem::get3DListenerAttributes (
    int listener,
    FMOD_VECTOR * pos,
    FMOD_VECTOR * vel,
    FMOD_VECTOR * forward,
    FMOD_VECTOR * up
);
```

## Parameters

*listener*

Listener ID in a multi-listener environment. Specify 0 if there is only 1 listener.

*pos*

Address of a variable that receives the position of the listener in world space, measured in distance units. Optional. Specify 0 to ignore.

*vel*

Address of a variable that receives the velocity of the listener measured in distance units **per second**. Optional. Specify 0 to ignore.

*forward*

Address of a variable that receives the forwards orientation of the listener. Optional. Specify 0 to ignore.

*up*

Address of a variable that receives the upwards orientation of the listener. Optional. Specify 0 to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable,

PlayStation 3, Wii

#### See Also

- [EventSystem::set3DListenerAttributes](#)
- [FMOD\\_VECTOR](#)

Version 4.12.03 Built on Feb 18, 2008

# EventSystem::get3DNumListeners

Retrieves the number of 3D listeners.?

## Syntax

```
FMOD_RESULT FModSystem::get3DNumListeners (
    int * numListeners
);
```

## Parameters

*numlisteners*

Address of a variable that receives the current number of 3D listeners in the 3D scene.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::set3DNumListeners](#)



# EventSystem::getCategory

Retrieve an event category object by name.?

## Syntax

```
FMOD_RESULT FModSystem::getCategory (
    const char * name,
    EventCategory ** category
);
```

## Parameters

*name*

The name of an event category within this event system. Specify "master" to retrieve the master event category.

*category*

Address of a variable to receive the selected event category within this event system.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Sub-categories can be retrieved by specifying their full path e.g. "vehicles/cars/racers".

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getCategoryByIndex](#)
- [EventSystem::getNumCategories](#)

# EventSystem::getCategoryByIndex

Retrieve an event category object by index.?

## Syntax

```
FMOD_RESULT FModSystem::getCategoryByIndex(  
    int index,  
    EventCategory ** category  
);
```

## Parameters

*index*

The index of an event category within this event system object. Indices are 0 based. Specify -1 to retrieve the master event category.

*category*

Address of a variable to receive the selected event category within this event system.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getCategory](#)
- [EventSystem::getNumCategories](#)

# EventSystem::getEvent

Retrieve an event object by name.?

## Syntax

```
FMOD_RESULT FModSystem::getEvent(  
    const char * name ,  
    FMODEVENTMODE mode ,  
    Event ** event  
);
```

## Parameters

*name*

The name of an event within this event system. Note: name must include full path including project name and any event group names e.g. "myproject/group1/group2/myevent"

*mode*

*event*

Address of a variable to receive the selected event within this event system.

## Return Values

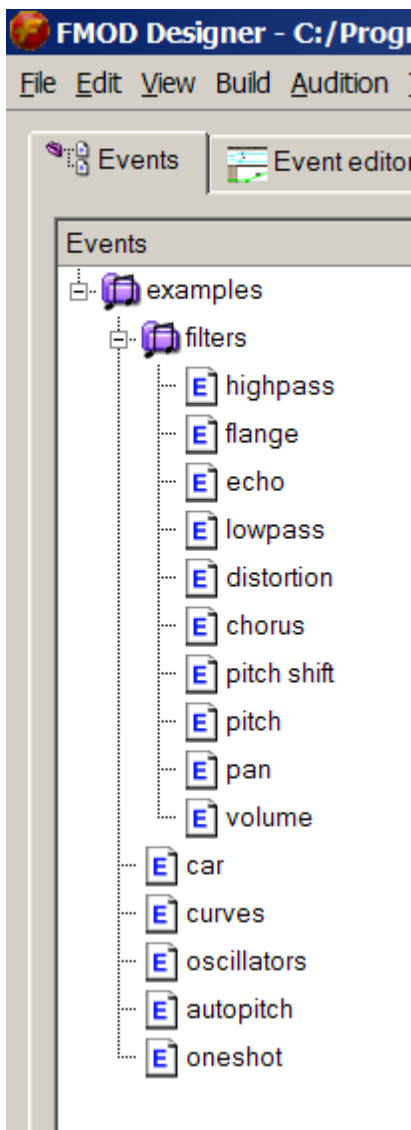
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event?

An event is the leaf of the event group tree. It is the actual sound to be played with complex behaviour designed by the sound designer.



In this case we are retrieving an event from an **event group**, so with the **"filters"** group we could get the echo event with **"echo"** as the name parameter.

If the programmer does not know which events are available, the sound designer tool can output a programmer report that lists the event group's events with the appropriate names and indices listed alongside them.

### Note!

- An event is retrieved from a pool of events (created earlier if FMOD\_EVENT\_CACHEEVENTS flag was set in [EventSystem::getGroup](#) / EventGroup::getGroup).
- Data may not be loaded from the disk for this event, so this event may trigger disk access. If you wish to pre-empt this use EventGroup::loadEventData first.
- The pool of events has a size determined by the 'max playbacks' property in the FMOD Designer tool in the event's property sheet.
- The pointer to will be getting will be a pointer to one of these event instances.
- If you call this function more times than there are event instances, then an invent handle may be stolen, or may fail. This behaviour also determined by the sound designer. The behaviour may be to steal the oldest event in the pool, steal the quietest event in the pool, or simply fail this getEvent and return null as the event handle.

### Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getGroup](#)
- [FMOD\\_EVENT\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008

# EventSystem::getEventBySystemID

Retrieve an event object by system wide unique identifier. All loaded events can be enumerated with this function and [EventSystem::getNumEvents](#).

## Syntax

```
FMOD_RESULT FMODSYS::getEventBySystemID(
    unsigned int systemid,
    FMODEVENTMODE mode,
    Event ** event
);
```

## Parameters

*systemid*

*mode*

*event*

Address of a variable to receive the selected event within this event system.

## Return Values

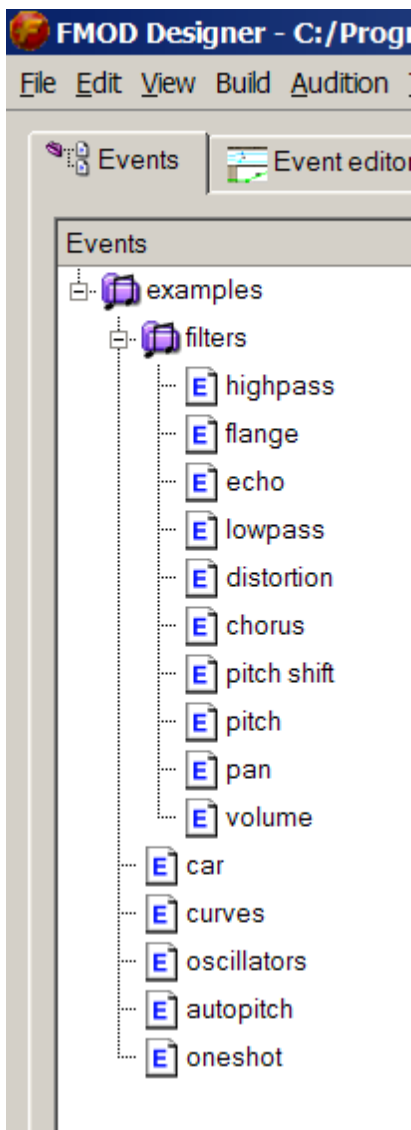
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event?

An event is the leaf of the event group tree. It is the actual sound to be played with complex behaviour designed by the sound designer.



In this case we are retrieving an event from an **event group**, so with the **"filters"** group we could get the echo event with **"echo"** as the name parameter.

If the programmer does not know which events are available, the sound designer tool can output a programmer report that lists the event group's events with the appropriate names and indices listed alongside them.

### Note!

- An event is retrieved from a pool of events (created earlier if FMOD\_EVENT\_CACHEEVENTS flag was set in EventSystem::getGroup / EventGroup::getGroup).
- Data may not be loaded from the disk for this event, so this event may trigger disk access. If you wish to pre-empt this use EventGroup::loadEventData first.
- The pool of events has a size determined by the 'max playbacks' property in the FMOD Designer tool in the event's property sheet.
- The pointer to will be getting will be a pointer to one of these event instances.
- If you call this function more times than there are event instances, then an invent handle may be stolen, or may fail. This behaviour also determined by the sound designer. The behaviour may be to steal the oldest event in the pool, steal the quietest event in the pool, or simply fail this getEvent and return null as the event handle.

### Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventSystem::getNumEvents](#)
- [FMOD\\_EVENT\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008



# EventSystem::getGroup

Retrieves an event group object by name.?

## Syntax

```
FMOD_RESULT FModSystem::getGroup(  
    const char * name ,  
    bool cacheEvents ,  
    EventGroup ** group  
);
```

## Parameters

*name*

The name of an event group that belongs to this event system. Note: name must include full path including project name and any event group names e.g. "myproject/group1/group2"

*cacheEvents*

If cacheEvents is true then all event instances within this event group will be pre-allocated so that there are no memory allocs when getEvent is called.

*group*

Address of a variable to receive the selected event group within this event system.

## Return Values

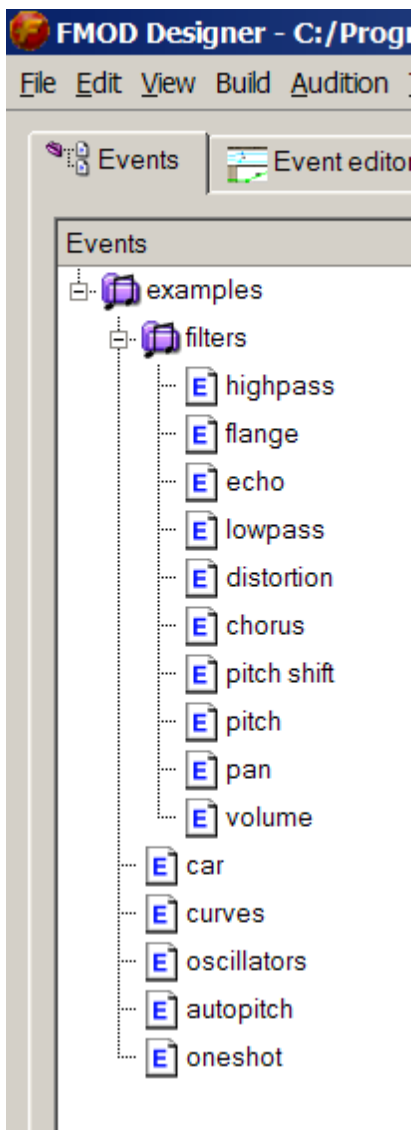
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

What is an event group?

An event group is a "folder" that stores events or sub-folders. With these folders a hierarchical tree can be built to store events in a more logical manner.



In this case we are retrieving an event group from another **event group**, so if this event group object was "**examples**" we could then get the event group "**filters**" with "**filters**" as the name parameter.

In this example "**filters**" is the only sub-group below "**examples**" so no other sub-groups are available here.

If the programmer does not know which sub-groups are available or which sub-group index matches which sub-group name, the sound designer tool can output a programmer report that lists the group's sub-groups with the appropriate names and indices listed alongside them.

The only benefit of retrieving an object by index is that it is slightly faster to do so than to retrieve it by name.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getGroup](#)
- [EventGroup::getGroupByIndex](#)



# EventSystem::getInfo

Retrieves information about the event system.?

## Syntax

```
FMOD_RESULT FMOD_System::getInfo (
    FMODEVENTSYSTEMINFO * info
);
```

## Parameters

*info*

Address of an [FMOD\\_EVENT\\_SYSTEMINFO](#) structure to receive event system information.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_EVENT\\_SYSTEMINFO](#)

# EventSystem::getNumCategories

Retrieve the number of categories for the event system.?

## Syntax

```
FMOD_RESULT FModSystem::getNumCategories (
    int * numcategories
);
```

## Parameters

*numcategories*

Address of a variable to receive the number of categories for this event system.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getCategoryByIndex](#)
- [EventSystem::getCategory](#)

# EventSystem::getNumEvents

Gets the total number of unique events loaded into the system at once, including those from different EventProjects. Mainly used for enumeration of all events loaded into the system. This can be done in conjunction with [EventSystem::getEventBySystemID](#).

## Syntax

```
FMOD_RESULT __stdcall EventSystem::getNumEvents (
    int * numevents
);
```

## Parameters

*numevents*

Pointer to a integer to retrieve the current number of unique events.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getEventBySystemID](#)

# EventSystem::getNumProjects

Retrieve the number of event projects within the top level event system.?

## Syntax

```
FMOD_RESULT FModEventSystem::getNumProjects (
    int *    num_projects
);
```

## Parameters

*numprojects*

Address of a variable to receive the number of event projects within the top level event system.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getProjectByIndex](#)

# EventSystem::getNumReverbPresets

Retrieve the number of reverb presets defined by the sound designer.?

## Syntax

```
FMOD_RESULT fmodSystem::getNumReverbPresets (
    int * numPresets
);
```

## Parameters

*numPresets*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use this in conjunction with [EventSystem::getReverbPresetByIndex](#) to enumerate all sound designer specified reverb presets.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getReverbPresetByIndex](#)
- [EventSystem::getReverbPreset](#)



# EventSystem::getProject

Retrieve an event project object by name.?

## Syntax

```
FMOD_RESULT FModSystem::getProject(  
    const char * name ,  
    EventProject ** project  
);
```

## Parameters

*name*

The name of an event project within this event system.

*project*

Address of a variable to receive the selected event project within this event system.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getProjectByIndex](#)
- [EventSystem::load](#)

# EventSystem::getProjectByIndex

Retrieve an event project object by index.?

## Syntax

```
FMOD_RESULT FModSystem::getProjectByIndex(  
    int index,  
    EventProject ** project  
);
```

## Parameters

*index*

The index of an event project within this event system object. Indices are 0 based.

*project*

Address of a variable to receive the selected event project within this event system.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getProject](#)
- [EventSystem::getNumProjects](#)
- [EventSystem::load](#)

# EventSystem::getReverbAmbientProperties

Retrieves the default reverb environment for the virtual reverb system.?

## Syntax

```
FMOD_RESULT __stdcall EventSystem::getReverbAmbientProperties (
    FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of a pointer to a [FMOD\\_REVERB\\_PROPERTIES](#) to receive the settings for the current ambient reverb setting.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

By default the ambient reverb is set to 'off'. This is the same as FMOD\_REVERB\_PRESET\_OFF.

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [EventSystem::setReverbAmbientProperties](#)
- [EventSystem::createReverb](#)

# EventSystem::getReverbPreset

Retrieves a reverb property structure containing a reverb preset created by the sound designer, by name.?

## Syntax

```
FMOD_RESULT EventSystem::getReverbPreset(  
    const char * name,  
    FMOD_REVERB_PROPERTIES * prop,  
    int * index  
);
```

## Parameters

*name*

The name of an event reverb within this event system.

*prop*

Address of a variable to receive a [FMOD\\_REVERB\\_PROPERTIES](#) containing the desired reverb preset.

*index*

Address of a variable to receive the index of the preset in the reverb list. Optional. This index is the index used in [EventSystem::getReverbPresetByIndex](#). Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use the retrieved [FMOD\\_REVERB\\_PROPERTIES](#) structure to pass to [EventSystem::setReverbProperties](#), [EventSystem::setReverbAmbientProperties](#) or [EventReverb::setProperties](#).

Reverb presets can also be retrieved with [EventSystem::getReverbPresetByIndex](#).

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [EventSystem::getReverbPresetByIndex](#)

- [EventSystem::setReverbProperties](#)
- [EventSystem::setReverbAmbientProperties](#)
- [EventReverb::setProperties](#)
- [EventReverb::release](#)

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# EventSystem::getReverbPresetByIndex

Retrieves a reverb property structure containing a reverb preset created by the sound designer, by index instead of name.?

## Syntax

```
FMOD_RESULT FModSystem::getReverbPresetByIndex(
    const int index,
    FMOD_REVERB_PROPERTIES * prop,
    char ** name
);
```

## Parameters

*index*

The index of an event reverb within this event system object. Indices are 0 based.

*prop*

Address of a variable to receive a [FMOD\\_REVERB\\_PROPERTIES](#) containing the desired reverb preset.

*name*

Address of a variable to receive a pointer to the name of the preset. Optional. This index is the name used in [EventSystem::getReverbPreset](#). Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration. Use the retrieved [FMOD\\_REVERB\\_PROPERTIES](#) structure to pass to EventSystem::setReverbProperties, EventSystem::setReverbAmbientProperties or EventReverb::setProperties.

## Remarks

All reverbs can be enumerated by using this function in conjunction with [EventSystem::getNumReverbPresets](#). Reverb presets can also be retrieved with [EventSystem::getReverbPreset](#).

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [EventSystem::getReverbPreset](#)
- [EventSystem::getNumReverbPresets](#)
- [EventReverb::release](#)

# EventSystem::getReverbProperties

Retrieves the current reverb environment.?

## Syntax

```
FMOD_RESULT fmodSystem::getReverbProperties (
    FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of a variable that receives the current reverb environment description.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::setReverbProperties](#)
- [Event::setReverbProperties](#)
- [Event::getReverbProperties](#)



# EventSystem::getSystemObject

Retrieve the event system's internal FMOD::System object for the low level FMOD Ex API.

## Syntax

```
FMOD_RESULT FModEx::getSystemObject(
    FMOD::System ** system
);
```

## Parameters

*system*

Address of a pointer to receive the FMOD::System pointer.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note! This should generally not be used unless you are trying to add features that the sound designer cannot provide! The aim of this API is to give the sound designer the control over the sound behaviour. If there are things missing from the EventSystem API that could be included contact FMOD support at [support@fmod.org](mailto:support@fmod.org) and it will be considered for addition.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

# EventSystem::getUserData

Retrieves the user value that that was set by calling the [EventSystem::setUserData](#) function.?

## Syntax

```
FMOD_RESULT FModSystem::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [EventSystem::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::setUserData](#)

# EventSystem::getVersion

Returns the current version of the event system being used.?

## Syntax

```
FMOD_RESULT FModSystem::getVersion(  
    unsigned int * version  
);
```

## Parameters

*version*

Address of a variable that receives the current event system version.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The version is a 32bit hexadecimal value formatted as 16:8:8, with the upper 16bits being the major version, the middle 8bits being the minor version and the bottom 8bits being the development version. For example a value of 00010106h is equal to 1.01.06.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::init](#)

# EventSystem::init

Initializes the event system object, FMOD system object and the sound device. This has to be called at the start of the user's program.

**?You must create an event system object with EventSystem\_Create.?**

## Syntax

```
FMOD_RESULT EventSystem::init(  
    int maxchannels,  
    FMOD_INIT_FLAGS flags,  
    void* extradata,  
    FMOD_EVENT_INIT_FLAGS eventflags  
);
```

## Parameters

*maxchannels*

The maximum number of channels to be used in FMOD. They are also called 'virtual channels' as you can play as many of these as you want, even if you only have a small number of hardware or software voices. See remarks for more.

*flags*

See [FMOD\\_INITFLAGS](#). This can be a selection of flags bitwise OR'ed together to change the behaviour of FMOD at initialization time.

*extradrivervdata*

Driver specific data that can be passed to the output plugin. For example the filename for the wav writer plugin. See [FMOD\\_OUTPUTTYPE](#) for what each output mode might take here. Optional. Specify 0 to ignore.

*eventflags*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

See FMOD Ex documentation for details on [FMOD\\_INITFLAGS](#) etc.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

#### See Also

- [\\_EventSystem\\_Create](#)

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# EventSystem::load

Loads an event file (.fev).?

## Syntax

```
FMOD_RESULT EventSystem::load(  
    const char * name_or_data ,  
    FMOD_EVENT_DATA_IN * data_in ,  
    EventProject * project  
);
```

## Parameters

*name\_or\_data*

Filename of the event file to be loaded or pointer to memory block if FMOD\_EVENT\_LOADINFO is provided and has a non-zero "loadfrommemory\_length" field.

*loadinfo*

Pointer to an FMOD\_EVENT\_LOADINFO structure which lets the user provide extended information about loading the file. Optional. Specify 0 or NULL to ignore.

*project*

Address of a variable to receive the specified project object. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Loading the event data file will not open any FSB files or allocate any memory for events. This is done with [EventGroup::loadEventData](#) (to load FSB data) and EventProject::getGroup / EventProject::getGroupByIndex / [EventGroup::getGroup](#) / [EventGroup::getGroupByIndex](#) to allocate memory for the event instances so that they can be played.

To load a .fev file from memory, pass a pointer to the memory block in "name\_or\_data" and provide an FMOD\_EVENT\_LOADINFO structure with the "loadfrommemory\_length" field set to the length of the memory block. The memory block can be freed immediately after this function returns.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::setMediaPath](#)
- [EventGroup::loadEventData](#)
- [EventSystem::getProject](#)
- [EventSystem::getProjectByIndex](#)
- [EventGroup::getGroup](#)
- [EventGroup::getGroupByIndex](#)

# EventSystem::registerMemoryFSB

For users that want to pre-load static sample FSB files into memory, this function can be used to stop FMOD from loading any FSB with the same filename from disk, and it will instead point to this memory instead.?

## Syntax

```
FMOD_RESULT EventSystem::registerMemoryFSB(
    const char * filename,
    void * fsbdata,
    unsigned int fsbdata_len,
    bool loadintoRSX
);
```

## Parameters

*filename*

File name that FMOD event system would use to load. FMOD will compare this string against media path + fsb filename or just fsb filename by itself.

*fsbdata*

Pointer to in memory FSB file.

*fsbdata\_len*

Length of in memory FSB file data in bytes.

*loadintoRSX*

PS3 only. When set to true, data pointed to by 'fsbdata' will be copied to the RSX memory pool specified at System::init.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When a bank is opened as streamed, these in memory FSB files are not referenced. If 'loadintoRSX' is set to true, data pointed to by 'fsbdata' will be copied to the RSX memory pool specified at System::init. It is safe to free 'fsbdata' when this is the case. NOTE: There is a performance penalty when using RSX memory. Please refer to the PS3 section of the FMOD documentation for more details.



## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventSystem::unregisterMemoryFSB](#)

Version 4.12.03 Built on Feb 18, 2008

# EventSystem::release

Closes and frees an event system object.?

## Syntax

```
FMOD_RESULT fmod::EventSystem::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This will free the event system object and everything created under it.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem\\_Create](#)
- [EventSystem::init](#)

# EventSystem::set3DListenerAttributes

This updates the position, velocity and orientation of the specified 3D sound listener.?

## Syntax

```
FMOD_RESULT EventSystem::set3DListenerAttributes (
    int listener,
    const FMOD_VECTOR * pos,
    const FMOD_VECTOR * vel,
    const FMOD_VECTOR * forward,
    const FMOD_VECTOR * up
);
```

## Parameters

*listener*

Listener ID in a multi-listener environment. Specify 0 if there is only 1 listener.

*pos*

Address of a variable that receives the position of the listener in world space, measured in distance units. You can specify 0 or NULL to not update the position.

*vel*

Address of a variable that receives the velocity of the listener measured in distance units **per second**. You can specify 0 or NULL to not update the velocity of the listener.

*forward*

Address of a variable that receives the forwards orientation of the listener. This vector must be of unit length and perpendicular to the up vector. You can specify 0 or NULL to not update the forwards orientation of the listener.

*up*

Address of a variable that receives the upwards orientation of the listener. This vector must be of unit length and perpendicular to the forwards vector. You can specify 0 or NULL to not update the upwards orientation of the listener.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

By default, FMOD uses a left-handed co-ordinate system. This means +X is right, +Y is up, and +Z is forwards. To change this to a right-handed coordinate system, use [FMOD\\_INIT\\_3D\\_RIGHTHANDED](#). This means +X is left, +Y is up, and +Z is forwards.

To map to another coordinate system, flip/negate and exchange these values.

Orientation vectors are expected to be of UNIT length. This means the magnitude of the vector should be 1.0.

A 'distance unit' is specified by the sound designer in the FMOD Designer tool. By default this is set to meters which is a distance scale of 1.0.

Always remember to use **units per second**, *not* units per frame as this is a common mistake and will make the doppler effect sound wrong.

For example, Do not just use (pos - lastpos) from the last frame's data for velocity, as this is not correct. You need to time compensate it so it is given in units per **second**.

You could alter your pos - lastpos calculation to something like this.

```
vel = (pos - lastpos) / time_taken_since_last_frame_in_seconds.
```

I.e. at 60fps the formula would look like this vel = (pos-lastpos) / 0.0166667.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::get3DListenerAttributes](#)
- [FMOD\\_INITFLAGS](#)
- [FMOD\\_VECTOR](#)

# EventSystem::set3DNumListeners

Sets the number of 3D 'listeners' in the 3D sound scene. This function is useful mainly for split-screen game purposes.?

## Syntax

```
FMOD_RESULT EventSystem::set3DNumListeners (
    int numListeners
);
```

## Parameters

*numlisteners*

Number of listeners in the scene. Valid values are from 1-4 inclusive. Default = 1.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If the number of listeners is set to more than 1, then panning and doppler are turned off. All sound effects will be mono.

FMOD uses a 'closest sound to the listener' method to determine what should be heard in this case.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::get3DNumListeners](#)
- [EventSystem::set3DListenerAttributes](#)

# EventSystem::setMediaPath

Specify a base search path for media files so they can be placed somewhere other than the directory of the main executable.?

## Syntax

```
FMOD_RESULT fmod::EventSystem::setMediaPath(  
    const char * path  
);
```

## Parameters

*path*

A character string containing a correctly formatted path to load media files from. NOTE: Must contain a trailing slash/backslash if filesystem requires it!

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

FMOD used to add a slash/backslash separator between the path provided here and the files that it needed to load. This caused inconsistent and inflexible behaviour so this function has been changed to expect a path that already contains a trailing slash/backslash. FMOD will no longer add any slash/backslash separators between the path specified here and the files that it needs to load. This allows the user to provide the correct separator for the filesystem in use - which may actually be the user's own filesystem which may or may not use a separator at all.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::load](#)

# EventSystem::setPluginPath

Specify a base search path for plugins so they can be placed somewhere other than the directory of the main executable.

## Syntax

```
FMOD_RESULT fmod::EventSystem::setPluginPath(  
    const char* path  
);
```

## Parameters

*path*

A character string containing a correctly formatted path to load plugins from. You can specify 0 or NULL to tell FMOD not to load any plugins.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::load](#)

# EventSystem::setReverbAmbientProperties

Sets a 'background' default reverb environment for the virtual reverb system. This is a reverb preset that will be morphed to if the listener is not within any virtual reverb zones.

?By default the ambient reverb is set to 'off'.?

## Syntax

```
FMOD_RESULT EventSystem::setReverbAmbientProperties (
    FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of a [FMOD\\_REVERB\\_PROPERTIES](#) structure containing the settings for the desired ambient reverb setting.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

There is one reverb DSP dedicated to providing a 3D reverb effect. This DSP's properties are a weighted sum of all the contributing virtual reverbs.

The default 3d reverb properties specify the reverb properties in the 3D volumes which has no virtual reverbs defined.

[EventSystem::getReverbPreset](#) and [EventSystem::getReverbPresetByIndex](#) can be used to retrieve sound designer defined presets, or it can be set programatically.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [EventSystem::getReverbAmbientProperties](#)
- [EventSystem::createReverb](#)



- [EventSystem::getReverbPreset](#)
- [EventSystem::getReverbPresetByIndex](#)

Version 4.12.03 Built on Feb 18, 2008

# EventSystem::setReverbProperties

Sets parameters for the global reverb environment.

?Reverb parameters can be set manually, or automatically using the pre-defined presets given in the fmod.h header.?

## Syntax

```
FMOD_RESULT EventSystem::setReverbProperties (
    const FMOD_REVERB_PROPERTIES * prop
);
```

## Parameters

*prop*

Address of an [FMOD\\_REVERB\\_PROPERTIES](#) structure which defines the attributes for the reverb.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

[EventSystem::getReverbPreset](#) and [EventSystem::getReverbPresetByIndex](#) can be used to retrieve sound designer defined presets, or it can be set programatically.

With [FMOD\\_HARDWARE](#) on Windows using EAX, the reverb will only work on [FMOD\\_3D](#) based sounds. [FMOD\\_SOFTWARE](#) does not have this problem and works on [FMOD\\_2D](#) and [FMOD\\_3D](#) based sounds.

On PlayStation 2, the reverb is limited to only a few reverb types that are not configurable. Use the FMOD\_PRESET\_PS2\_XXX presets.

On Xbox, it is possible to apply reverb to [FMOD\\_2D](#) and [FMOD\\_HARDWARE](#) based voices using this function. By default reverb is turned off for [FMOD\\_2D](#) hardware based voices.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_REVERB\\_PROPERTIES](#)
- [EventSystem::getReverbProperties](#)

- [Event::setReverbProperties](#)
- [Event::getReverbProperties](#)
- [EventSystem::getReverbPreset](#)
- [EventSystem::getReverbPresetByIndex](#)

Version 4.12.03 Built on Feb 18, 2008

# EventSystem::setUserData

Sets a user value that the EventSystem object will store internally. Can be retrieved with [EventSystem::getUserData](#)?

## Syntax

```
FMOD_RESULT EventSystem::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the EventSystem object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [EventSystem::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getUserData](#)

# EventSystem::unload

Unloads all loaded event projects?

## Syntax

```
FMOD_RESULT FModEventSystem::unload();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::load](#)

# EventSystem::unregisterMemoryFSB

De-register an in memory FSB from the system.?

## Syntax

```
FMOD_RESULT EventSystem::unregisterMemoryFSB(  
    const char * filename  
);
```

## Parameters

*filename*

FSB filename to unregister.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Any references to this filename by the EventSystem internally will be loaded from disk as normal if this FSB is unregistered as an in memory FSB. If 'loadintorsx' was used in [EventSystem::registerMemoryFSB](#), this data is freed from the RSX pool.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::registerMemoryFSB](#)

# EventSystem::update

Updates the event system. This should be called once per 'game' tick, or once per frame in your application.?

## Syntax

```
FMOD_RESULT fmod::EventSystem::update();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventSystem::init](#)

# EventProject Interface

[EventProject::getEvent](#)

[EventProject::getEventByProjectID](#)

[EventProject::getGroup](#)

[EventProject::getGroupByIndex](#)

[EventProject::getInfo](#)

[EventProject::getNumEvents](#)

[EventProject::getNumGroups](#)

[EventProject::getUserData](#)

[EventProject::release](#)

[EventProject::setUserData](#)



# EventProject::getEvent

Retrieve a an event object by name.?

## Syntax

```
FMOD_RESULT EventProject::getEvent(  
    const char * name ,  
    FMODEVENTMODE mode ,  
    Event ** event  
);
```

## Parameters

*name*

The name of an event within this event project. Note: name must include full path including any event group names e.g. "group1/group2/myevent"

*mode*

*event*

Address of a variable to receive the selected event within this event project.

## Return Values

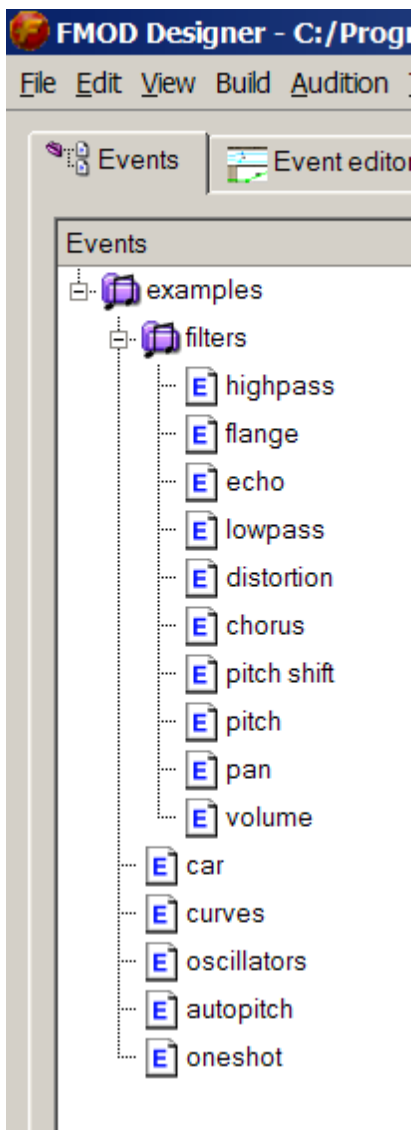
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event?

An event is the leaf of the event group tree. It is the actual sound to be played with complex behaviour designed by the sound designer.



In this case we are retrieving an event from an **event group**, so with the **"filters"** group we could get the echo event with **"echo"** as the name parameter.

If the programmer does not know which events are available, the sound designer tool can output a programmer report that lists the event group's events with the appropriate names and indices listed alongside them.

### Note!

- An event is retrieved from a pool of events (created earlier if FMOD\_EVENT\_CACHEEVENTS flag was set in [EventSystem::getGroup](#) / [EventGroup::getGroup](#)).
- Data may not be loaded from the disk for this event, so this event may trigger disk access. If you wish to pre-empt this use EventGroup::loadEventData first.
- The pool of events has a size determined by the 'max playbacks' property in the FMOD Designer tool in the event's property sheet.
- The pointer to will be getting will be a pointer to one of these event instances.
- If you call this function more times than there are event instances, then an event handle may be stolen, or may fail. This behaviour also determined by the sound designer. The behaviour may be to steal the oldest event in the pool, steal the quietest event in the pool, or simply fail this getEvent and return null as the event handle.

### Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::getEventByProjectID](#)
- [EventGroup::getEventByIndex](#)
- [EventSystem::getGroup](#)
- [EventGroup::getGroup](#)
- [FMOD\\_EVENT\\_MODE](#)

# EventProject::getEventByProjectID

Retrieve an event handle by a project unique id.?

## Syntax

```
FMOD_RESULT EventProject::getEventByProjectID(
    unsigned int projectid,
    FMODEVENTMODE mode,
    Event ** event
);
```

## Parameters

*projectid*

The project id of an event within this event project. Unique ids can be found in the programmer report generated when the project is built, and the C header.

*mode*

*event*

Address of a variable to receive the selected event within this event project.

## Return Values

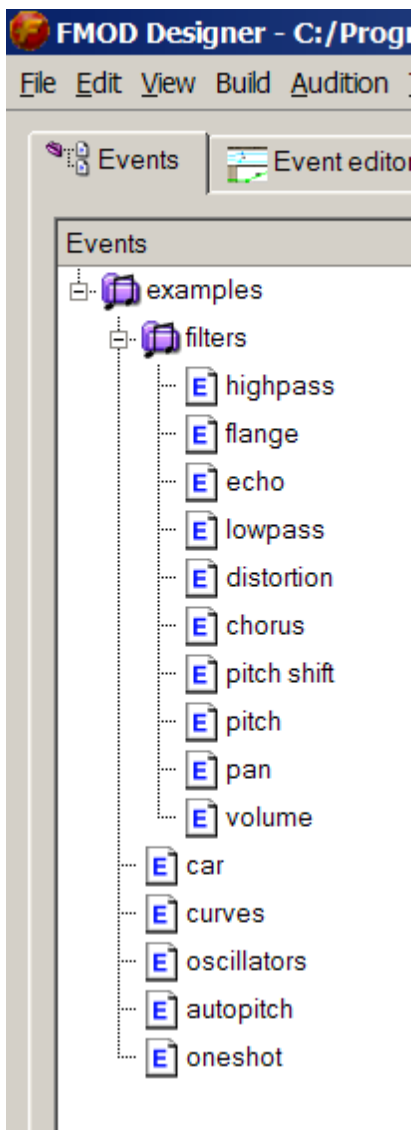
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event?

An event is the leaf of the event group tree. It is the actual sound to be played with complex behaviour designed by the sound designer.



In this case we are retrieving an event from an **event group**, so with the **"filters"** group we could get the echo event with **"echo"** as the name parameter.

If the programmer does not know which events are available, the sound designer tool can output a programmer report that lists the event group's events with the appropriate names and indices listed alongside them.

### Note!

- An event is retrieved from a pool of events (created earlier if FMOD\_EVENT\_CACHEEVENTS flag was set in [EventSystem::getGroup](#) / [EventGroup::getGroup](#)).
- Data may not be loaded from the disk for this event, so this event may trigger disk access. If you wish to pre-empt this use EventGroup::loadEventData first.
- The pool of events has a size determined by the 'max playbacks' property in the FMOD Designer tool in the event's property sheet.
- The pointer to will be getting will be a pointer to one of these event instances.
- If you call this function more times than there are event instances, then an event handle may be stolen, or may fail. This behaviour also determined by the sound designer. The behaviour may be to steal the oldest event in the pool, steal the quietest event in the pool, or simply fail this getEvent and return null as the event handle.

### Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::getEvent](#)
- [EventProject::getNumEvents](#)
- [EventGroup::getEventByIndex](#)
- [EventSystem::getGroup](#)
- [EventGroup::getGroup](#)
- [FMOD\\_EVENT\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008

# EventProject::getGroup

Retrieves an event group object by name.?

## Syntax

```
FMOD_RESULT EventProject::getGroup(  
    const char * name ,  
    bool cache_events ,  
    EventGroup ** group  
);
```

## Parameters

*name*

The name of an event group that belongs to this event project.

*cacheevents*

If cacheevents is true then all event instances within this event group will be pre-allocated so that there are no memory allocs when getEvent is called.

*group*

Address of a variable to receive the selected event group within this event project.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::getGroupByIndex](#)
- [EventGroup::getGroup](#)
- [EventGroup::getGroupByIndex](#)

# EventProject::getGroupByIndex

Retrieves an event group object by index.?

## Syntax

```
FMOD_RESULT FMOD_EVENTPROJECT::getGroupByIndex(
    int index,
    bool cache_events,
    EventGroup ** group
);
```

## Parameters

*index*

The index of an event group within this event project. Indices are 0 based.

*cache\_events*

If cache\_events is true then all event instances within this event group will be pre-allocated so that there are no memory allocs when getEvent is called.

*group*

Address of a variable to receive the selected event group within this event project.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::getNumGroups](#)
- [EventProject::getGroup](#)
- [EventGroup::getGroup](#)
- [EventGroup::getGroupByIndex](#)





# EventProject::getInfo

Retrieve information about this event project.?

## Syntax

```
FMOD_RESULTE e ntPro ge t: ge t i nfo (  
  i nt * i nd e x,  
  c h a r ** name  
) ;
```

## Parameters

*index*

Address of a variable to receive the event project index.

*name*

Address of a variable to receive the event project name.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getInfo](#)
- [Event::getInfo](#)

# EventProject::getNumEvents

Returns the total number of events for this project only.?

## Syntax

```
FMOD_RESULT FMOD_EVENTPROJECT::getNumEvents (
    int * numevents
);
```

## Parameters

*numevents*

Address of a variable to receive the number of events in this project.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::getEventByProjectID](#)

# EventProject::getNumGroups

Retrieves the number of event groups stored within this event project.?

## Syntax

```
FMOD_RESULT FModEventProject::getNumGroups (
    int * numgroups
);
```

## Parameters

*numgroups*

Address of a variable to receive the number of groups within this event project.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::getGroupByIndex](#)

# EventProject::getUserData

Retrieves the user value that that was set by calling the [EventProject::setUserData](#) function.?

## Syntax

```
FMOD_RESULT EventProject::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [EventProject::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::setUserData](#)

# EventProject::release

Release this event project and all the events/eventgroups that it contains.?

## Syntax

```
FMOD_RESULT EventProject::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventSystem::getProject](#)
- [\\_EventSystem::getProjectByIndex](#)

# EventProject::setUserData

Sets a user value that the EventProject object will store internally. Can be retrieved with [EventProject::getUserData](#)?

## Syntax

```
FMOD_RESULT EventProject::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the EventProject object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [EventProject::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventProject::getUserData](#)

# EventGroup Interface

[EventGroup::freeEventData](#)

[EventGroup::getEvent](#)

[EventGroup::getEventByIndex](#)

[EventGroup::getGroup](#)

[EventGroup::getGroupByIndex](#)

[EventGroup::getInfo](#)

[EventGroup::getNumEvents](#)

[EventGroup::getNumGroups](#)

[EventGroup::getNumProperties](#)

[EventGroup::getParentGroup](#)

[EventGroup::getParentProject](#)

[EventGroup::getProperty](#)

[EventGroup::getPropertyByIndex](#)

[EventGroup::getState](#)

[EventGroup::getUserData](#)

[EventGroup::loadEventData](#)

[EventGroup::setUserData](#)



# EventGroup::freeEventData

Free the resources for an EventGroup and all subgroups under it or for just a single specified event.?

## Syntax

```
FMOD_RESULT EventGroup::freeEventData (
    Event * event,
    bool waituntilready
);
```

## Parameters

*event*

Single event to free resources for. Specify 0 or NULL to ignore.

*waituntilready*

If TRUE, this function will block until all pending asynchronous loads have completed before freeing the event data. If FALSE, this function will return [FMOD\\_ERR\\_NOTREADY](#) if any asynchronous loads are pending and it will NOT free any event data. Default = TRUE.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

If no event is specified then resources for all events in this EventGroup, and all EventGroups within this EventGroup, will be freed.

If an event is specified then just the resources for that event will be freed.

NOTE: This function does not completely remove events from memory, it simply frees any resources allocated by them. To completely remove events from memory, use EventSystem::unload.

Use waituntilready = false in time-critical situations to avoid blocking the main thread. Note that if [FMOD\\_ERR\\_NOTREADY](#) is returned from this function then no event data was actually freed - you will need to call this function again until it succeeds.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::loadEventData](#)
- [EventSystem::getGroup](#)

Version 4.12.03 Built on Feb 18, 2008

# EventGroup::getEvent

Retrieve a an event object by name.?

## Syntax

```
FMOD_RESULT FmodEventGroup::getEvent(  
    const char * name ,  
    FMODEVENTMODE mode ,  
    Event ** event  
);
```

## Parameters

*name*

The name of an event within this event group.

*mode*

*event*

Address of a variable to receive the selected event within this event group.

## Return Values

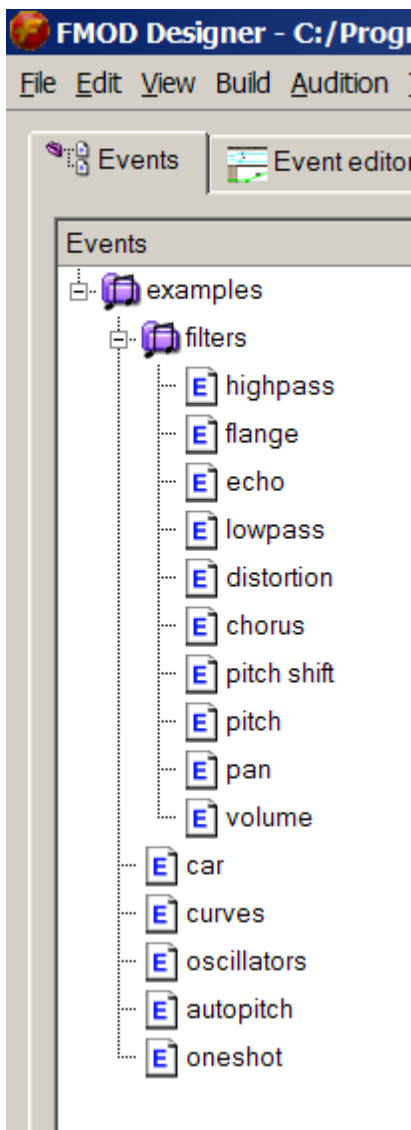
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event?

An event is the leaf of the event group tree. It is the actual sound to be played with complex behaviour designed by the sound designer.



In this case we are retrieving an event from an **event group**, so with the **"filters"** group we could get the echo event with **"echo"** as the name parameter.

If the programmer does not know which events are available, the sound designer tool can output a programmer report that lists the event group's events with the appropriate names and indices listed alongside them.

### Note!

- An event is retrieved from a pool of events (created earlier if FMOD\_EVENT\_CACHEEVENTS flag was set in [EventSystem::getGroup](#) / [EventGroup::getGroup](#)).
- Data may not be loaded from the disk for this event, so this event may trigger disk access. If you wish to pre-empt this use EventGroup::loadEventData first.
- The pool of events has a size determined by the 'max playbacks' property in the FMOD Designer tool in the event's property sheet.
- The pointer to will be getting will be a pointer to one of these event instances.
- If you call this function more times than there are event instances, then an invent handle may be stolen, or may fail. This behaviour also determined by the sound designer. The behaviour may be to steal the oldest event in the pool, steal the quietest event in the pool, or simply fail this getEvent and return null as the event handle.

### Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getEventByIndex](#)
- [EventSystem::getGroup](#)
- [EventGroup::getGroup](#)
- [FMOD\\_EVENT\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008

# EventGroup::getEventByIndex

Retrieve an event object by index for this group.?

## Syntax

```
FMOD_RESULT fmod_event_group_get_event_by_index(  
    int index,  
    FMODEVENTMODE mode,  
    Event **event  
);
```

## Parameters

*index*

The index of an event within this event sub-group. Indices are 0 based.

*mode*

*event*

Address of a variable to receive the selected event within this event group.

## Return Values

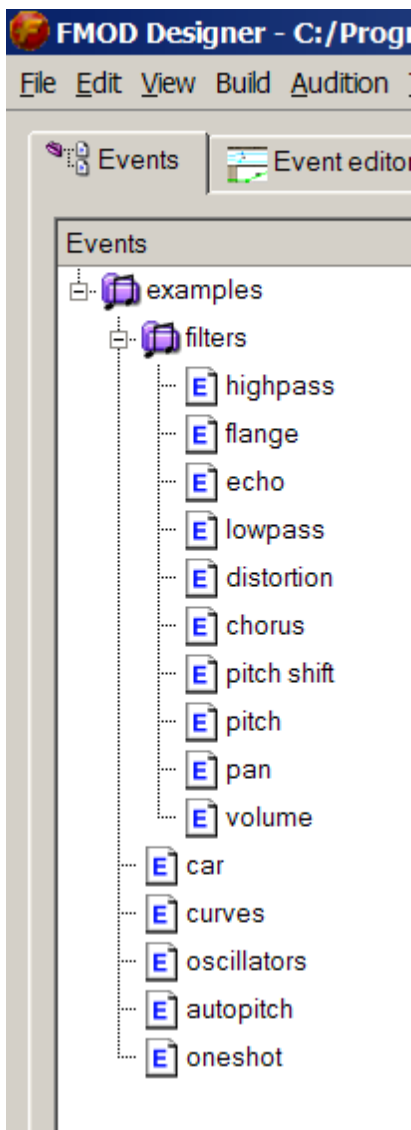
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event?

An event is the leaf of the event group tree. It is the actual sound to be played with complex behaviour designed by the sound designer.



In this case we are retrieving an event from an **event group**, so with the **"filters"** group we could get the echo event with **2** as the index parameter.

If the programmer does not know which events are available or which event index matches which event name, the sound designer tool can output a programmer report that lists the event group's events with the appropriate names and indices listed alongside them.

The only benefit of retrieving an object by index is that it is slightly faster to do so than to retrieve it by name.

#### **Note!**

- An event is retrieved from a pool of events (created earlier if FMOD\_EVENT\_CACHEEVENTS flag was set in EventSystem::getGroup / EventGroup::getGroup).
- Data may not be loaded from the disk for this event, so this event may trigger disk access. If you wish to pre-empt this use EventGroup::loadEventData first.
- The pool of events has a size determined by the 'max playbacks' property in the FMOD Designer tool in the event's property sheet.
- The pointer to will be getting will be a pointer to one of these event instances.
- If you call this function more times than there are event instances, then an event handle may be stolen, or may fail. This behaviour also determined by the sound designer. The behaviour may be to steal the oldest event in the pool, steal the quietest event in the pool, or simply fail this getEvent and return null as the event handle.

## **Platforms Supported**

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

**See Also**

- [EventGroup::getEvent](#)
- [EventGroup::getNumEvents](#)
- [FMOD\\_EVENT\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008



# EventGroup::getGroup

Retrieves an event group's sub-group object by name.?

## Syntax

```
FMOD_RESULT FMOD_EVENTGROUP::getGroup(  
    const char * name ,  
    bool cache_events ,  
    FMOD_EVENTGROUP ** group  
);
```

## Parameters

*name*

The name of an event sub-group that belongs to this event group.

*cacheevents*

If cacheevents is true then all event instances within this event group will be pre-allocated so that there are no memory allocs when getEvent is called.

*group*

Address of a variable to receive the selected event sub-group within this event group.

## Return Values

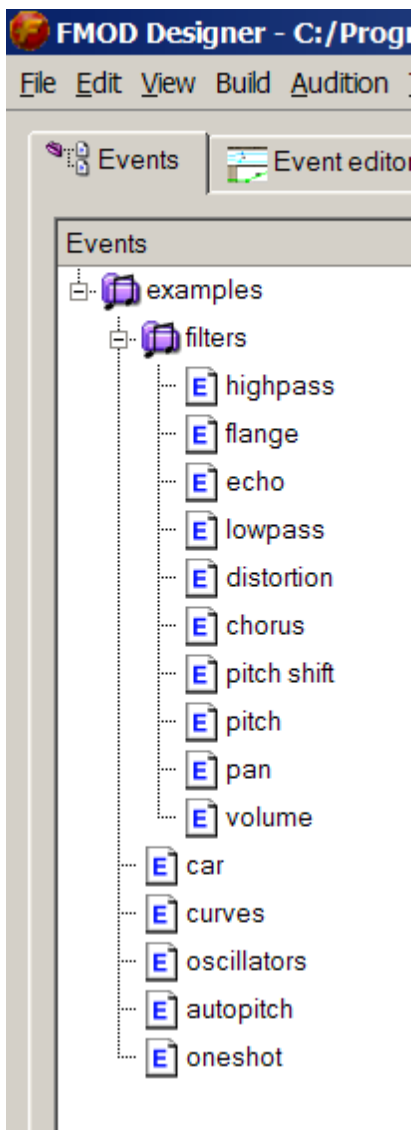
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event group?

An event group is a "folder" that stores events or sub-folders. With these folders a hierarchical tree can be built to store events in a more logical manner.



In this case we are retrieving an event group from another **event group**, so if this event group object was "**examples**" we could then get the event group "**filters**" with "**filters**" as the name parameter.

In this example "**filters**" is the only sub-group below "**examples**" so no other sub-groups are available here.

If the programmer does not know which sub-groups are available or which sub-group index matches which sub-group name, the sound designer tool can output a programmer report that lists the group's sub-groups with the appropriate names and indices listed alongside them.

The only benefit of retrieving an object by index is that it is slightly faster to do so than to retrieve it by name.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getGroup](#)
- [EventGroup::getGroupByIndex](#)



# EventGroup::getGroupByIndex

Retrieves an event group's sub-group object by index.?

## Syntax

```
FMOD_RESULT FMOD_EVENTGROUP::getGroupByIndex(  
    int index,  
    bool cache_events,  
    EventGroup ** group  
);
```

## Parameters

*index*

The index of an event sub-group within this event group. Indices are 0 based.

*cache\_events*

If cache\_events is true then all event instances within this event group will be pre-allocated so that there are no memory allocs when getEvent is called.

*group*

Address of a variable to receive the selected event sub-group within this event group.

## Return Values

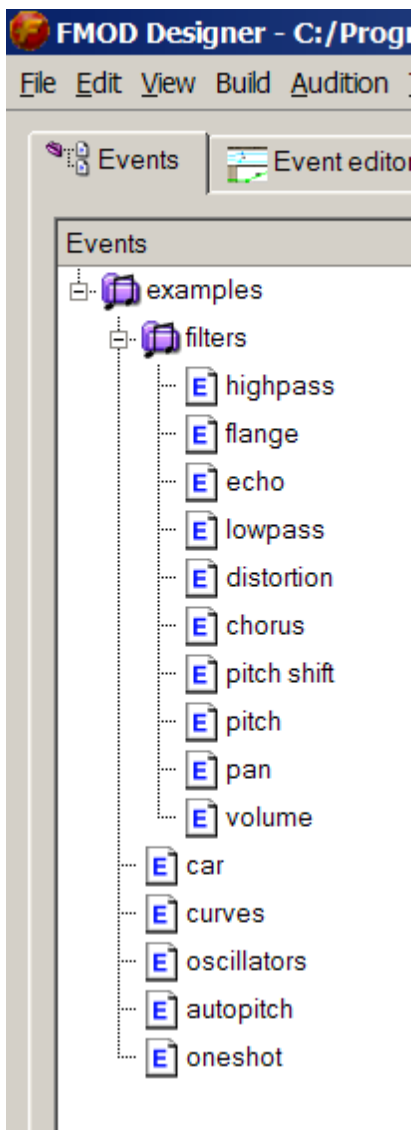
If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

### What is an event group?

An event group is a "folder" that stores events or sub-folders. With these folders a hierarchical tree can be built to store events in a more logical manner.



In this case we are retrieving an event group from another **event group**, so if this event group object was "**examples**" we could then get the event group "**filters**" with **0** as the index parameter.

In this example "**filters**" is the only sub-group below "**examples**" so no other sub-groups are available here.

If the programmer does not know which groups are available or which event group index matches which group name, the sound designer tool can output a programmer report that lists the group's sub-groups with the appropriate names and indices listed alongside them.

The only benefit of retrieving an object by index is that it is slightly faster to do so than to retrieve it by name.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getGroup](#)
- [EventGroup::getGroup](#)
- [EventGroup::getNumGroups](#)



# EventGroup::getInfo

Retrieve information about this event group.?

## Syntax

```
FMOD_RESULTE e ne de pount pount index (
    i nt * i ndex,
    c har ** name
);
```

## Parameters

*index*

Address of a variable to receive the event group index.

*name*

Address of a variable to receive the event group name.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventParameter::getInfo](#)
- [Event::getInfo](#)

# EventGroup::getNumEvents

Retrieves the number of event events stored within this event group.?

## Syntax

```
FMOD_RESULT fmod_event_group::getNumEvents (
    int * numevents
);
```

## Parameters

*numevents*

Adress of a variable to receive the number of events within this event group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getEventByIndex](#)



# EventGroup::getNumGroups

Retrieves the number of event groups stored within this event group.?

## Syntax

```
FMOD_RESULT fmod_event_group::getNumGroups (
    int * numgroups
);
```

## Parameters

*numgroups*

Adress of a variable to receive the number of groups within this event group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getGroupByIndex](#)

# EventGroup::getNumProperties

Retrieve the number of properties for an event group.?

## Syntax

```
FMOD_RESULT fmod_event_group_get_num_properties (
    int * num_properties
);
```

## Parameters

*numproperties*

Address of a variable to receive the number of properties for this event group.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getPropertyByIndex](#)

# EventGroup::getParentGroup

Retrieves the eventgroup object to which this eventgroup belongs.?

## Syntax

```
FMOD_RESULT eventgroup::getParentGroup(
    EventGroup ** group
);
```

## Parameters

*group*

Address of a variable that receives a pointer to the eventgroup's parent eventgroup

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The top level eventgroup will return a parent of 0 or NULL, as it has no parent.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getGroup](#)
- [EventGroup::getGroupByIndex](#)

# EventGroup::getParentProject

Retrieves the eventproject object to which this eventgroup belongs.?

## Syntax

```
FMOD_RESULT fmod_eventgroup::getParentProject(
    EventProject ** project
);
```

## Parameters

*project*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getGroup](#)
- [EventGroup::getGroupByIndex](#)

# EventGroup::getProperty

Retrieve an event group property by name.?

## Syntax

```
FMOD_RESULT fmod_event_group::getProperty (
    const char * property name ,
    void * value
);
```

## Parameters

*propertyname*

Name of the property to retrieve. This is the name that was specified in FMOD Designer.

*value*

Address of a variable to receive the event group property.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getPropertyByIndex](#)
- [EventGroup::getNumProperties](#)
- [EventGroup::getEvent](#)

# EventGroup::getPropertyByIndex

Retrieve an event group property by index.?

## Syntax

```
FMOD_RESULT fmod_event_group::getPropertyByIndex(  
    int propertyindex,  
    void * value  
);
```

## Parameters

*propertyindex*

Index of the property to retrieve.

*value*

Address of a variable to receive the event group property.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getProperty](#)
- [EventGroup::getNumProperties](#)
- [EventGroup::getEvent](#)

# EventGroup::getState

Retrieves the current state of an event group.?

## Syntax

```
FMOD_RESULT fmod_event_group_get_state (
    FMOD_EVENT_STATE * state
);
```

## Parameters

*state*

Address of a variable that receives the event group's current state.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

When [FMOD\\_EVENT\\_STATE\\_PLAYING](#) is true, at least one event in the group is playing. When [FMOD\\_EVENT\\_STATE\\_PLAYING](#) is false, no event in the group is playing.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_EVENT\\_STATE](#)

# EventGroup::getUserData

Retrieves the user value that that was set by calling the [EventGroup::setUserData](#) function.?

## Syntax

```
FMOD_RESULT fmod_event_group_get_userdata (
    void ** userdata
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [EventGroup::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::setUserData](#)



# EventGroup::loadEventData

Loads the resources for all events within an event group.?

## Syntax

```
FMOD_RESULT EventGroup::loadEventData (
    FMOD_EVENT_RESOURCE resource,
    FMOD_EVENT_MODE mode
);
```

## Parameters

*resource*

Type of data to load. Either load samples, streams, or both. See [FMOD\\_EVENT\\_RESOURCE](#).

*mode*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use [EventGroup::freeEventData](#) to unload sample banks and close streams under this group.

Note that if another event in a different group is still using the sound data, it will not be freed until those events have had freeEventData called on them as well. (On their parent group).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::freeEventData](#)
- [EventSystem::getGroup](#)
- [FMOD\\_EVENT\\_RESOURCE](#)
- [FMOD\\_EVENT\\_MODE](#)



# EventGroup::setUserData

Sets a user value that the EventGroup object will store internally. Can be retrieved with [EventGroup::getUserData](#)?

## Syntax

```
FMOD_RESULT fmod_event_group_set_userdata (
    void * userdata
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the EventGroup object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [EventGroup::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getUserData](#)

# Event Interface

[Event::get3DAttributes](#)  
[Event::get3DOcclusion](#)  
[Event::getCategory](#)  
[Event::getChannelGroup](#)  
[Event::getInfo](#)  
[Event::getMute](#)  
[Event::getNumParameters](#)  
[Event::getNumProperties](#)  
[Event::getParameter](#)  
[Event::getParameterByIndex](#)  
[Event::getParentGroup](#)  
[Event::getPaused](#)  
[Event::getPitch](#)  
[Event::getProperty](#)  
[Event::getPropertyByIndex](#)  
[Event::getReverbProperties](#)  
[Event::getState](#)  
[Event::getUserData](#)  
[Event::getVolume](#)  
[Event::set3DAttributes](#)  
[Event::set3DOcclusion](#)  
[Event::setCallback](#)  
[Event::setMute](#)  
[Event::setPaused](#)  
[Event::setPitch](#)  
[Event::setProperty](#)  
[Event::setPropertyByIndex](#)  
[Event::setReverbProperties](#)  
[Event::setUserData](#)  
[Event::setVolume](#)  
[Event::start](#)  
[Event::stop](#)

# Event::get3DAttributes

Retrieves the position and velocity of an event.?

## Syntax

```
FMOD_RESULT Event::get3DAttributes (
    FMOD_VECTOR * position,
    FMOD_VECTOR * velocity,
    FMOD_ORIENTATION * orientation
);
```

## Parameters

*position*

Address of a variable that receives the position in 3D space of the event. Optional. Specify 0 to ignore.

*velocity*

Address of a variable that receives the velocity in 'distance units per second' in 3D space of the event. See remarks. Optional. Specify 0 to ignore.

*orientation*

Address of a variable that receives the orientation of the event. Optional. Specify 0 to ignore. Only used for events with sound cones specified.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A 'distance unit' is specified in the FMOD Designer tool and are the distance units used by the game (i.e. feet, meters, inches, centimeters etc). An event has to be 3D to have its 3d position and velocity set.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::set3DAttributes](#)

Version 4.12.03 Built on Feb 18, 2008

# Event::get3DOcclusion

Retrieves the the EAX or software based occlusion factors for an event.?

## Syntax

```
FMOD_RESULT event::get3DOcclusion(  
    float * directOcclusion,  
    float * reverbOcclusion  
);
```

## Parameters

*directOcclusion*

Address of a variable that receives the occlusion factor for a voice for the direct path. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0. Optional. Specify 0 or NULL to ignore.

*reverbOcclusion*

Address of a variable that receives the occlusion factor for a voice for the reverb mix. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0. Optional. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

With EAX based sound cards, or I3DL2 based hardware accelerated voices, this will attenuate the sound and frequencies. With non EAX or I3DL2 hardware accelerated voices, then the volume is attenuated by the directOcclusion factor.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::set3DOcclusion](#)





# Event::getCategory

Retrieve an event category that this event belongs to.?

## Syntax

```
FMOD_RESULT event::getCategory (
    EventCategory ** category
);
```

## Parameters

*category*

Address of a variable to receive the event category.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getInfo](#)
- [EventGroup::getEvent](#)

# Event::getChannelGroup

Retrieves a pointer to a lower level ChannelGroup class, mainly so that the programmer can add a custom DSP effect with ChannelGroup::addDSP.?

## Syntax

```
FMOD_RESULT event::getChannelGroup(
    FMOD_CHANNELGROUP ** channelgroup
);
```

## Parameters

*channelgroup*

Address of a variable to receive a pointer to a low level ChannelGroup class.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

# Event::getInfo

Retrieves information about the event.?

## Syntax

```
FMOD_RESULT Event::getInfo (
    int * index,
    char ** name,
    FMOD_EVENT_INFO * info
);
```

## Parameters

*index*

Address of a variable to receive the event group's index. Specify 0 or NULL to ignore.

*name*

Address of a variable to receive the event name. Specify 0 or NULL to ignore.

*info*

Address of an [FMOD\\_EVENT\\_INFO](#) structure to receive extended event information. Specify 0 or NULL to ignore.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The [FMOD\\_EVENT\\_INFO](#) structure has members that need to be initialized before Event::getInfo is called. Always initialize the [FMOD\\_EVENT\\_INFO](#) structure before calling Event::getInfo!

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getEvent](#)
- [EventGroup::getEventByIndex](#)

- [EventParameter::getInfo](#)
- [FMOD\\_EVENT\\_INFO](#)

Version 4.12.03 Built on Feb 18, 2008

# Event::getMute

Retrieves the muted state of an event.?

## Syntax

```
FMOD_RESULT Event::getMute (
    bool * mute
);
```

## Parameters

*mute*

Address of a variable to receive the muted state of the event.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::setMute](#)

# Event::getNumParameters

Retrieve the number of parameters for an event.?

## Syntax

```
FMOD_RESULT Event::getNumParameters (
    int * numParameters
);
```

## Parameters

*numParameters*

Address of a variable to receive the number of parameters for this event.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getParameterByIndex](#)

# Event::getNumProperties

Retrieve the number of properties for an event.?

## Syntax

```
FMOD_RESULT event::getNumProperties (
    int * numproperties
);
```

## Parameters

*numproperties*

Address of a variable to receive the number of properties for this event.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getPropertyByIndex](#)

# Event::getParameter

Retrieve an event parameter object by name.?

## Syntax

```
FMOD_RESULT Event::getParameter(  
    const char * name ,  
    EventParameter ** parameter  
);
```

## Parameters

*name*

The name of an event parameter that belongs to this event.

*parameter*

Address of a variable to receive the selected event parameter within this event.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Each event will have a list of parameters to control the playback behaviour of the event. For example, if a sound designer made a car engine event, one of the parameters might be 'RPM'.

If the programmer does not know which parameters are available, the sound designer tool can output a programmer report that lists the event's parameters with the appropriate names and indices listed alongside them.

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getParameterByIndex](#)
- [EventGroup::getEvent](#)
- [EventGroup::getEventByIndex](#)



- [FMOD\\_EVENT\\_MODE](#)

Version 4.12.03 Built on Feb 18, 2008

# Event::getParameterByIndex

Retrieve an event parameter by index.?

## Syntax

```
FMOD_RESULT Event::getParameterByIndex(  
    int index,  
    EventParameter ** parameter  
);
```

## Parameters

*index*

The index of an event parameter within this event. Indices are 0 based. Pass -1 to retrieve this event's primary parameter.

*parameter*

Address of a variable to receive the event parameter object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Each event will have a list of parameters to control the playback behaviour of the event. For example, if a sound designer made a car engine event, one of the parameters might be 'RPM'.

If the programmer does not know which parameters are available or which index matches which parameter, the sound designer tool can output a programmer report that lists the event's parameters with the appropriate names and indices listed alongside them.

The only benefit of retrieving a parameter by index is that it is slightly faster to do so than to retrieve it by name.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getParameter](#)
- [Event::getNumParameters](#)
- [EventGroup::getEvent](#)

- [EventGroup::getEventByIndex](#)

Version 4.12.03 Built on Feb 18, 2008

# Event::getParentGroup

Retrieves the eventgroup object to which this event belongs.?

## Syntax

```
FMOD_RESULT event::getParentGroup(  
    EventGroup ** group  
);
```

## Parameters

*group*

Address of a variable that receives a pointer to the event's parent eventgroup

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::getEvent](#)
- [EventGroup::getEventByIndex](#)

# Event::getPaused

Retrieves the paused state of an event.?

## Syntax

```
FMOD_RESULT event::getPaused(  
    bool * paused  
);
```

## Parameters

*paused*

Address of a variable to receive the paused state of the event.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::setPaused](#)

# Event::getPitch

Retrieves the overall pitch of an event.?

## Syntax

```
FMOD_RESULT Event::getPitch(
    float * pitch,
    FMOD_EVENT_PITCHUNITS units
);
```

## Parameters

*pitch*

Address of a variable to receive the current pitch level of the event. 0.0 = normal pitch (default).

*units*

The desired units for the retrieved pitch value.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::setPitch](#)
- [FMOD\\_EVENT\\_PITCHUNITS](#)

# Event::getProperty

Retrieve an event property by name.?

## Syntax

```
FMOD_RESULT Event::getProperty (
    const char * propertyName,
    void * value,
    bool this_instance
);
```

## Parameters

*propertyName*

Name of the property to retrieve. This is the name that was specified in FMOD Designer.

*value*

Address of a variable to receive the event property.

*this\_instance*

If TRUE then retrieve the per-instance property value, if FALSE then retrieve the parent ([FMOD\\_EVENT\\_INFOONLY](#)) event's property value.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A parent, or [FMOD\\_EVENT\\_INFOONLY](#), event is the prototype that all its event instances are based on. An event instance obtained by using any of the `getEventXXX` functions will be initialized with the current property values of its parent event. After an event instance is obtained, its property values may be modified using [Event::setProperty](#) and [Event::setPropertyByIndex](#) so that they differ from their parent event's properties. Use the 'this\_instance' parameter to specify whether to retrieve the property value of the parent event or the specific event instance.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_Event::getPropertyByIndex](#)
- [\\_Event::setProperty](#)
- [\\_Event::setPropertyByIndex](#)
- [\\_EventGroup::getEvent](#)



# Event::getPropertyByIndex

Retrieve an event property by index.?

## Syntax

```
FMOD_RESULT Event::getPropertyByIndex(
    int propertyindex,
    void * value,
    bool this_instance
);
```

## Parameters

*propertyindex*

Index of the property to retrieve. See [FMOD\\_EVENT\\_PROPERTY](#) for details.

*value*

Address of a variable to receive the event property.

*this\_instance*

If TRUE then retrieve the per-instance property value, if FALSE then retrieve the parent ([FMOD\\_EVENT\\_INFOONLY](#)) event's property value.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A parent, or [FMOD\\_EVENT\\_INFOONLY](#), event is the prototype that all its event instances are based on. An event instance obtained by using any of the `getEventXXX` functions will be initialized with the current property values of its parent event. After an event instance is obtained, its property values may be modified using [Event::setProperty](#) and [Event::setPropertyByIndex](#) so that they differ from their parent event's properties. Use the 'this\_instance' parameter to specify whether to retrieve the property value of the parent event or the specific event instance.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_Event::getProperty](#)
- [\\_Event::getNumProperties](#)
- [\\_Event::setProperty](#)
- [\\_Event::setPropertyByIndex](#)
- [\\_EventGroup::getEvent](#)
- [FMOD\\_EVENT\\_PROPERTY](#)

# Event::getReverbProperties

Retrieves the current reverb properties for this event.?

## Syntax

```
FMOD_RESULT Event::getReverbProperties (
    FMOD_REVERB_CHANNELPROPERTIES * prop
);
```

## Parameters

*prop*

Address of a variable to receive the FSOUND\_REVERB\_CHANNELPROPERTIES information.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Xbox, Xbox360, PlayStation 2, PlayStation 3

## See Also

- [Event::setReverbProperties](#)
- [FMOD\\_REVERB\\_CHANNELPROPERTIES](#)

# Event::getState

Retrieves the current state of an event.?

## Syntax

```
FMOD_RESULT event::getState (
    FMOD_EVENT_STATE * state
);
```

## Parameters

*state*

Address of a variable that receives the event's current state.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::loadEventData](#)
- [EventGroup::getEvent](#)
- [FMOD\\_EVENT\\_STATE](#)
- [FMOD\\_EVENT\\_MODE](#)

# Event::getUserData

Retrieves the user value that that was set by calling the [Event::setUserData](#) function.?

## Syntax

```
FMOD_RESULT event::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [Event::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::setUserData](#)

# Event::getVolume

Retrieves the overall volume of an event.?

## Syntax

```
FMOD_RESULT Event::getVolume (
    float * volume
);
```

## Parameters

*volume*

Address of a variable to receive the current volume level of the event. 0.0 = silent, 1.0 = full volume (default).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::setVolume](#)

# Event::set3DAttributes

Sets the 3d position and velocity of an event.?

## Syntax

```
FMOD_RESULT event::set3DAttributes (
    const FMOD_VECTOR * position,
    const FMOD_VECTOR * velocity,
    const FMOD_QUIET * orientation
);
```

## Parameters

### *position*

Position in 3D space of the event. Specifying 0 / null will ignore this parameter.

### *velocity*

Velocity in 'distance units per second' in 3D space of the event. See remarks. Specifying 0 / null will ignore this parameter.

### *orientation*

Orientation of the event sound cone. Only used if the event has a cone specified to determine cone detection, otherwise just specify 0 / null. Specifying 0 / null will ignore this parameter.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A 'distance unit' is specified in the FMOD Designer tool and are the distance units used by the game (i.e. feet, meters, inches, centimeters etc). An event has to be 3D to have its 3d position and velocity set.

Before getting an event with 'just fail if quietest' max playbacks behaviour, this function should be called on an EVENT\_INFOONLY event to allow the event system to estimate volume. The various getEvent\* functions copy the 3D attributes from the EVENT\_INFOONLY event to the event that is returned, so it is not necessary to set the 3D attributes again after getting a real event.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_Event::get3DAttributes](#)



# Event::set3DOcclusion

Sets the EAX or software based occlusion factors for an event. This function can be called to produce the same audible effects as the FMOD geometry engine, just without the built in polygon processing.

## Syntax

```
FMOD_RESULT Event::set3DOcclusion(
    float directOcclusion,
    float reverbOcclusion
);
```

## Parameters

*directOcclusion*

Occlusion factor for a voice for the direct path. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0.

*reverbOcclusion*

Occlusion factor for a voice for the reverb mix. 0.0 = not occluded. 1.0 = fully occluded. Default = 0.0.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function does not go through and overwrite the channel occlusion factors, it calls [ChannelGroup::set3DOcclusion](#). This means that final occlusion values will be affected by both Event occlusion and geometry (if any).

With EAX based sound cards, or I3DL2 based hardware accelerated voices, this will attenuate the sound and frequencies. With non EAX or I3DL2 hardware accelerated voices, then the volume is attenuated by the directOcclusion factor.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::get3DOcclusion](#)
- [Event::setReverbProperties](#)
- [ChannelGroup::set3DOcclusion](#)

Version 4.12.03 Built on Feb 18, 2008

# Event::setCallback

Sets a callback so that when certain event behaviours happen, they can be caught by the user.?

## Syntax

```
FMOD_RESULT event::setCallback (
    FMOD_EVENT_CALLBACK callback,
    void * userdata
);
```

## Parameters

*callback*

Pointer to a callback to be called by FMOD.

*userdata*

Userdata pointer to be passed to callback.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

wav sync points are supported.

These can be created by placing 'markers' in the original source wavs using a tool such as Sound Forge or Cooledit. Callbacks will be automatically generated when these markers are encountered.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_EVENT\\_CALLBACK](#)
- [FMOD\\_EVENT\\_CALLBACKTYPE](#)



# Event::setMute

Mutes or unmutes an event for runtime reasons.?

## Syntax

```
FMOD_RESULT Event::setMute (
    bool mute
);
```

## Parameters

*mute*

Mute state of the event. true = muted, false = unmuted.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

An event can have several hardware/software voices playing under it at once so this function mutes all relevant voices for this event.

This function is not to be used unless needed for runtime reasons.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_Event::getMute](#)

# Event::setPaused

Pauses or unpauses an event for runtime reasons.?

## Syntax

```
FMOD_RESULT Event::setPaused(  
    bool paused  
);
```

## Parameters

*paused*

Paused state of the event. true = paused, false = unpaused.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

An event can have several hardware/software voices playing under it at once so this function pauses all relevant voices for this event.

This function is not to be used unless needed for runtime reasons.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getPaused](#)

# Event::setPitch

Sets the overall pitch of an event.?

## Syntax

```
FMOD_RESULT Event::setPitch(
    float pitch,
    FMOD_EVENT_PITCHUNITS units
);
```

## Parameters

*pitch*

Pitch level of the event. 0.0 = normal pitch (default).

*units*

The units in which the new pitch level is specified.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

An event can have several hardware/software voices playing under it at once so this function scales all relevant voice pitches for this event.

This function is not to be used unless needed for runtime reasons, as the sound designer will have set the appropriate event pitch level in the FMOD Designer tool.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getPitch](#)
- [FMOD\\_EVENT\\_PITCHUNITS](#)





# Event::setProperty

Set an event property by name.?

## Syntax

```
FMOD_RESULT Event::setProperty (
    const char * propertyName,
    void * value,
    bool this_instance
);
```

## Parameters

*propertyName*

Name of the property to set. This is the name that was specified in FMOD Designer.

*value*

Pointer to the new value for this event property.

*this\_instance*

If TRUE then set the per-instance property value, if FALSE then set the property value of all event instances and also the parent ([FMOD\\_EVENT\\_INFOONLY](#)) event.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A parent, or [FMOD\\_EVENT\\_INFOONLY](#), event is the prototype that all its event instances are based on. An event instance obtained by using any of the `getEventXXX` functions will be initialized with the current property values of its parent event. After an event instance is obtained, its property values may be modified using `Event::setProperty` and [Event::setPropertyByIndex](#) so that they differ from their parent event's properties. Use the 'this\_instance' parameter to specify whether to set the property value of all event instances and also the parent event or just the specific event instance.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getProperty](#)
- [Event::setPropertyByIndex](#)
- [EventGroup::getEvent](#)

Version 4.12.03 Built on Feb 18, 2008

# Event::setPropertyByIndex

Set an event property by index.?

## Syntax

```
FMOD_RESULT Event::setPropertyByIndex(  
    int propertyIndex,  
    void * value,  
    bool this_instance  
);
```

## Parameters

*propertyindex*

Index of the property to set. See [FMOD\\_EVENT\\_PROPERTY](#) for details.

*value*

Pointer to the new value for this event property.

*this\_instance*

If TRUE then set the per-instance property value, if FALSE then set the property value of all event instances and also the parent ([FMOD\\_EVENT\\_INFOONLY](#)) event.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

A parent, or [FMOD\\_EVENT\\_INFOONLY](#), event is the prototype that all its event instances are based on. An event instance obtained by using any of the `getEventXXX` functions will be initialized with the current property values of its parent event. After an event instance is obtained, its property values may be modified using [Event::setProperty](#) and [Event::setPropertyByIndex](#) so that they differ from their parent event's properties. Use the 'this\_instance' parameter to specify whether to set the property value of all event instances and also the parent event or just the specific event instance.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getProperty](#)
- [Event::setProperty](#)
- [EventGroup::getEvent](#)
- [FMOD\\_EVENT\\_PROPERTY](#)

# Event::setReverbProperties

Sets the event specific reverb properties for sounds created with [FMOD\\_HARDWARE](#), including things like wet/dry mix (room size), and things like obstruction and occlusion properties.?

## Syntax

```
FMOD_RESULT Event::setReverbProperties (
    const FMOD_REVERB_CHANNELPROPERTIES * prop
);
```

## Parameters

*prop*

Pointer to a [FMOD\\_REVERB\\_CHANNELPROPERTIES](#) structure definition.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Under Win32 / Win64, you must be using [FMOD\\_OUTPUTTYPE\\_DSOUND](#) as the output mode for this to work. In DSOUND mode, the reverb will only work if you have an EAX compatible soundcard such as the Sound Blaster, and your sound was created with the [FMOD\\_HARDWARE](#) and [FMOD\\_3D](#) flags.

On PlayStation2, the 'Room' parameter is the only parameter supported. The hardware only allows 'on' or 'off', so the reverb will be off when 'Room' is -10000 and on for every other value.

On Xbox, it is possible to apply reverb to [FMOD\\_2D](#) based voices using this function. By default reverb is turned off for [FMOD\\_2D](#) based voices.

NOTE: This function overrides values set with [Event::set3DOcclusion](#) and also overrides the "Reverb Level" property defined using the FMOD Designer tool. If you need [Event::set3DOcclusion](#) or "3D Reverb Level" functionality then factor it into your [FMOD\\_REVERB\\_CHANNELPROPERTIES](#) values.

## Platforms Supported

Win32, Win64, Xbox, Xbox360, PlayStation 2, PlayStation 3

## See Also

- [Event::getReverbProperties](#)
- [System::setReverbProperties](#)

- [FMOD\\_REVERB\\_CHANNELPROPERTIES](#)
- [Event::set3DOcclusion](#)

Version 4.12.03 Built on Feb 18, 2008

# Event::setUserData

Sets a user value that the Event object will store internally. Can be retrieved with [Event::getUserData](#)?

## Syntax

```
FMOD_RESULT Event::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the Event object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [Event::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getUserData](#)

# Event::setVolume

Sets the overall volume of an event.?

## Syntax

```
FMOD_RESULT Event::setVolume (
    float volume
);
```

## Parameters

*volume*

Volume level of the event. 0.0 = silent, 1.0 = full volume (default).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

An event can have several hardware/software voices playing under it at once so this function scales all relevant voice volumes for this event.

This function is not to be used unless needed for runtime reasons, as the sound designer will have set the appropriate event volume level in the FMOD Designer tool.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::getVolume](#)



# Event::start

Start this event playing.?

## Syntax

```
FMOD_RESULT Event::start();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Call [Event::stop](#) to halt playback of an event.

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::stop](#)
- [EventGroup::getEvent](#)
- [EventGroup::getEventByIndex](#)
- [FMOD\\_EVENT\\_MODE](#)

# Event::stop

Stop this event playing.?

## Syntax

```
FMOD_RESULT Event::stop(  
    bool immediate  
);
```

## Parameters

*immediate*

Set this to true to force the event to stop immediately, ignoring the "Fadeout time" property

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::start](#)
- [EventGroup::getEvent](#)
- [EventGroup::getEventByIndex](#)
- [FMOD\\_EVENT\\_MODE](#)

# EventParameter Interface

[EventParameter::getInfo](#)  
[EventParameter::getRange](#)  
[EventParameter::getSeekSpeed](#)  
[EventParameter::getUserData](#)  
[EventParameter::getValue](#)  
[EventParameter::getVelocity](#)  
[EventParameter::keyOff](#)  
[EventParameter::setSeekSpeed](#)  
[EventParameter::setUserData](#)  
[EventParameter::setValue](#)  
[EventParameter::setVelocity](#)

# EventParameter::getInfo

Retrieve information about this event parameter.?

## Syntax

```
FMOD_RESULT eventParameter::getInfo (
    int * index,
    char ** name
);
```

## Parameters

*index*

Address of a variable to receive the event parameters index into the parent event.

*name*

Address of a variable to receive the event parameter name.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Mainly used for display purposes, this function returns a pointer to memory containing the event's name. Do not modify or try to free this memory.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::getRange](#)

# EventParameter::getRange

Retrieve the minimum and maximum values for this event parameter.?

## Syntax

```
FMOD_RESULT FMOD_EventParameter::getRange (
    float * rangemin,
    float * rangemax
);
```

## Parameters

*rangemin*

Address of variable to receive the minimum value allowed for this EventParameter.

*rangemax*

Address of variable to receive the maximum value allowed for this EventParameter.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This parameter is defined by the sound designer, and usually has a logical meaning, such as RPM for a car engine for example.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::setValue](#)
- [EventParameter::getValue](#)
- [EventParameter::getInfo](#)



# EventParameter::getSeekSpeed

Receives the seek velocity of an event.?

## Syntax

```
FMOD_RESULT EventParameter::getSeekSpeed(
    float * value
);
```

## Parameters

*value*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::setSeekSpeed](#)

# EventParameter::getUserData

Retrieves the user value that that was set by calling the [EventParameter::setUserData](#) function.?

## Syntax

```
FMOD_RESULT eventparameter::getuserdata (
    void ** userdata
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [EventParameter::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventParameter::setUserData](#)



# EventParameter::getValue

Retrieve the current value of this parameter.?

## Syntax

```
FMOD_RESULT eventParameter::getValue (
    float * value
);
```

## Parameters

*value*

Address of variable to receive the parameter value.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This parameter is defined by the sound designer, and has a minimum and maximum value. It usually has a logical meaning, such as RPM for a car engine for example.

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::setValue](#)
- [EventParameter::getRange](#)
- [EventParameter::getInfo](#)

# EventParameter::getVelocity

Receives the velocity of an event.?

## Syntax

```
FMOD_RESULT EventParameter::getVelocity (
    float * value
);
```

## Parameters

*value*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::setVelocity](#)

# EventParameter::keyOff

Triggers a keyoff on an event parameter that has sustain points in it. If an event parameter is currently sustaining on a sustain point, triggering a keyoff will release it and allow the parameter to continue.

## Syntax

```
FMOD_RESULT FMOD_EVENT_PARAMETER::keyOff();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Keyoffs can be triggered in advance of a sustain point being reached, so that they continue past the sustain point ahead of time.

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::setVelocity](#)
- [EventParameter::getVelocity](#)

# EventParameter::setSeekSpeed

Sets the seek velocity of a parameter.?

## Syntax

```
FMOD_RESULT EventParameter::setSeekSpeed(
    float value
);
```

## Parameters

*value*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note that currently setting the velocity of an event parameter will set the velocity for all instances of this event. This value is normally set by the sound designer but may be used if the programmer wishes to vary it.

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::getSeekSpeed](#)

# EventParameter::setUserData

Sets a user value that the EventParameter object will store internally. Can be retrieved with [EventParameter::getUserData](#)?

## Syntax

```
FMOD_RESULT EventParameter::setUserData (
    void * userData
);
```

## Parameters

*userData*

Address of user data that the user wishes stored within the EventParameter object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [EventParameter::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventParameter::getUserData](#)

# EventParameter::setValue

Set the 'value' of this parameter.?

## Syntax

```
FMOD_RESULT eventParameter::setValue (
    float value
);
```

## Parameters

*value*

Value to set this parameter to. Note! Must lie in the range described by [EventParameter::getRange](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This parameter is defined by the sound designer, and has a minimum and maximum value. It usually has a logical meaning, such as RPM for a car engine for example.

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::getValue](#)
- [EventParameter::getRange](#)

# EventParameter::setVelocity

Sets the velocity of a parameter. In most cases the velocity of a parameter will be 0, unless it is a time based event.?

## Syntax

```
FMOD_RESULT FMOD_EVENT_PARAMETER::setVelocity (
    float value
);
```

## Parameters

*value*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Note that currently setting the velocity of an event parameter will set the velocity for all instances of this event. This value is normally set by the sound designer but may be used if the programmer wishes to pause or speed up / slow down the parameter movement.

Attempting to use this function on an [FMOD\\_EVENT\\_INFOONLY](#) event will cause an [FMOD\\_ERR\\_INVALID\\_HANDLE](#) error to be returned.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventParameter::getVelocity](#)

# EventCategory Interface

[EventCategory::getCategory](#)  
[EventCategory::getCategoryByIndex](#)  
[EventCategory::getChannelGroup](#)  
[EventCategory::getEventByIndex](#)  
[EventCategory::getInfo](#)  
[EventCategory::getMute](#)  
[EventCategory::getNumCategories](#)  
[EventCategory::getNumEvents](#)  
[EventCategory::getPaused](#)  
[EventCategory::getPitch](#)  
[EventCategory::getUserData](#)  
[EventCategory::getVolume](#)  
[EventCategory::setMute](#)  
[EventCategory::setPaused](#)  
[EventCategory::setPitch](#)  
[EventCategory::setUserData](#)  
[EventCategory::setVolume](#)  
[EventCategory::stopAllEvents](#)



# EventCategory::getCategory

Retrieve an event category object by name.?

## Syntax

```
FMOD_RESULT F_CALLBACK getCategory (
    const char * name,
    EventCategory ** category
);
```

## Parameters

*name*

The name of an event category that belongs to this event category.

*category*

Address of a variable to receive the selected event category within this event category.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::getCategoryByIndex](#)
- [EventSystem::getCategory](#)

# EventCategory::getCategoryByIndex

Retrieve an event category object by index.?

## Syntax

```
FMOD_RESULT F_CALLBACK getCategoryByIndex(  
    int index,  
    EventCategory ** category  
);
```

## Parameters

*index*

The index of an event category within this event category. Indices are 0 based.

*category*

Address of a variable to receive the event category object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::getCategory](#)
- [EventSystem::getCategory](#)

# EventCategory::getChannelGroup

Retrieves a pointer to a lower level ChannelGroup class, mainly so that the programmer can add a custom DSP effect with ChannelGroup::addDSP.?

## Syntax

```
FMOD_RESULT F_CALLBACK EventCategory::getChannelGroup(
    FMOD_CHANNELGROUP ** channelgroup
);
```

## Parameters

*channelgroup*

Address of a variable to receive a pointer to a low level ChannelGroup class.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

# EventCategory::getEventByIndex

Retrieve an event object by index.?

## Syntax

```
FMOD_RESULT F_CALLBACK EventCategory::getEventByIndex(
    int index,
    FMODEVENTMODE *eventmode,
    Event **event
);
```

## Parameters

*index*

The index of an event within this event category. Indices are 0 based.

*eventmode*

*event*

Address of a variable to receive the event object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getCategory](#)
- [EventCategory::getNumEvents](#)

# EventCategory::getInfo

Retrieve information about this event category.?

## Syntax

```
FMOD_RESULT EventCategory::getInfo (
    int * index,
    char ** name
);
```

## Parameters

*index*

Address of a variable to receive the event category's index.

*name*

Address of a variable to receive the event category's name.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Mainly used for display purposes, this function returns a pointer to memory containing the event category's name. Do not modify or try to free this memory.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getCategory](#)

# EventCategory::getMute

Retrieves the mute state of an event category.?

## Syntax

```
FMOD_RESULT FMOD_EventCategory::getMute (
    bool * mute
);
```

## Parameters

*mute*

Address of a variable to receive the mute state of the event category.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventCategory::setMute](#)

# EventCategory::getNumCategories

Retrieve the number of sub-categories below this event category.?

## Syntax

```
FMOD_RESULT F_CALLBACK EventCategory::getNumCategories (
    int * numcategories
);
```

## Parameters

*numcategories*

Address of a variable to receive the number of categories in this category.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::getCategory](#)
- [EventSystem::getCategory](#)

# EventCategory::getNumEvents

Retrieve the number of events within this event category.?

## Syntax

```
FMOD_RESULT EventCategory::getNumEvents (
    int * numevents
);
```

## Parameters

*numevents*

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getCategory](#)
- [EventCategory::getEventByIndex](#)



# EventCategory::getPaused

Retrieves the paused state of an event category.?

## Syntax

```
FMOD_RESULT fmodEventCategory::getPaused(  
    bool * paused  
);
```

## Parameters

*paused*

Address of a variable to receive the paused state of the event category.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::setPaused](#)

# EventCategory::getPitch

Retrieves the overall pitch of an event category.?

## Syntax

```
FMOD_RESULT FMOD_EventCategory::getPitch(
    float * pitch,
    FMOD_EVENT_PITCHUNITS units
);
```

## Parameters

*pitch*

Address of a variable to receive the current pitch level of the event category. 0.0 = normal pitch (default).

*units*

The desired units for the retrieved pitch value.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::setPitch](#)
- [FMOD\\_EVENT\\_PITCHUNITS](#)

# EventCategory::getUserData

Retrieves the user value that that was set by calling the [EventCategory::setUserData](#) function.?

## Syntax

```
FMOD_RESULT FMOD_EVENT_CATEGORY::getUserData (
    void ** userData
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [EventCategory::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::setUserData](#)

# EventCategory::getVolume

Retrieves the overall volume of an event category.?

## Syntax

```
FMOD_RESULT FMOD_EVENT_CATEGORY::getVolume (
    float * volume
);
```

## Parameters

*volume*

Address of a variable to receive the current volume level of the event category. 0.0 = silent, 1.0 = full volume (default).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::setVolume](#)

# EventCategory::setMute

Pauses or unpauses an event category for runtime reasons.?

## Syntax

```
FMOD_RESULT FMOD_EventCategory::setMute (
    bool mute
);
```

## Parameters

*mute*

Mute the event category. true = muted, false = unmuted.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventCategory::getMute](#)

# EventCategory::setPaused

Pauses or unpauses an event category for runtime reasons.?

## Syntax

```
FMOD_RESULT fmodEventCategory::setPaused(  
    bool paused  
);
```

## Parameters

*paused*

Paused state of the event category. true = paused, false = unpaused.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventCategory::getPaused](#)

# EventCategory::setPitch

Sets the overall pitch of an event category.?

## Syntax

```
FMOD_RESULT fmod_event_category::setPitch(  
    float pitch,  
    FMOD_EVENT_PITCHUNITS units  
);
```

## Parameters

*pitch*

Pitch level of the event category. 0.0 = normal pitch (default).

*units*

The units in which the new pitch level is specified.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::getPitch](#)
- [FMOD\\_EVENT\\_PITCHUNITS](#)

# EventCategory::setUserData

Sets a user value that the EventCategory object will store internally. Can be retrieved with [EventCategory::getUserData](#).

## Syntax

```
FMOD_RESULT EventCategory::setUserData (
    void * userData
);
```

## Parameters

*userData*

Address of user data that the user wishes stored within the EventCategory object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [EventCategory::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::getUserData](#)



# EventCategory::setVolume

Sets the overall volume of an event category.?

## Syntax

```
FMOD_RESULT FmodEventCategory::setVolume (
    float volume
);
```

## Parameters

*volume*

Volume level of the event category. 0.0 = silent, 1.0 = full volume (default).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

An event category can have several events playing under it at once so this function scales all relevant event volumes for this event category.

This function is not to be used unless needed for runtime reasons, as the sound designer will have set the appropriate event category volume level in the FMOD Designer tool.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventCategory::getVolume](#)

# EventCategory::stopAllEvents

Stops all events in this category and subcategories.?

## Syntax

```
FMOD_RESULT fmodEventCategory::stopAllEvents();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

# EventReverb Interface

[EventReverb::get3DAttributes](#)

[EventReverb::getActive](#)

[EventReverb::getProperties](#)

[EventReverb::getUserData](#)

[EventReverb::release](#)

[EventReverb::set3DAttributes](#)

[EventReverb::setActive](#)

[EventReverb::setProperties](#)

[EventReverb::setUserData](#)

# EventReverb::get3DAttributes

Gets the 3D attributes of the event 3D reverb object?

## Syntax

```
FMOD_RESULT EventReverb::get3DAttributes (
    FMOD_VECTOR * position,
    float * mindistance,
    float * maxdistance
);
```

## Parameters

*position*

pointer to a vector in 3D space where the reverb is centred

*mindistance*

radius within which the reverb has full effect

*maxdistance*

radius outside of which the reverb has zero effect

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 3D reverb object is a sphere having 3D attributes (position, minimum distance, maximum distance) and reverb properties.

The properties and 3D attributes of all reverb objects collectively determine, along with the listener's position, the settings of and input gains into a single 3D reverb DSP.

Please note that this only applies to software channels. When the listener is within the sphere of effect of one or more 3d reverbs, the listener's 3D reverb properties are a weighted combination of such 3d reverbs. When the listener is outside all of the reverbs, the 3D reverb setting is set to the default ambient reverb setting.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventReverb::set3DAttributes](#)

Version 4.12.03 Built on Feb 18, 2008

# EventReverb::getActive

Retrieves the active state of the reverb object.?

## Syntax

```
FMOD_RESULT EventReverb::getActive (
    bool * active
);
```

## Parameters

*active*

Address of a variable to receive the current active state of the reverb object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventReverb::setActive](#)
- [EventSystem::createReverb](#)

# EventReverb::getProperties

Retrieves the current reverb properties for this event 3d reverb object.?

## Syntax

```
FMOD_RESULT eventReverb::getProperties (
    FMOD_REVERB_PROPERTIES * props
);
```

## Parameters

*props*

Address of a variable to receive the [FMOD\\_REVERB\\_PROPERTIES](#) information.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventReverb::setProperties](#)
- [FMOD\\_REVERB\\_PROPERTIES](#)

# EventReverb::getUserData

Retrieves the user value that that was set by calling the [EventReverb::setUserData](#) function.?

## Syntax

```
FMOD_RESULT fmod_reverb::get_user_data (
    void ** user_data
);
```

## Parameters

*userdata*

Address of a pointer that receives the data specified with the [EventReverb::setUserData](#) function.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventReverb::setUserData](#)



# EventReverb::release

Release memory for an event reverb object.?

## Syntax

```
FMOD_RESULT EventReverb::release();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This will release this event reverb object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::getReverbPreset](#)
- [EventSystem::getReverbPresetByIndex](#)

# EventReverb::set3DAttributes

Sets the 3D attributes of the event 3D reverb object?

## Syntax

```
FMOD_RESULT EventReverb::set3DAttributes (
    const FMOD_VECTOR * position,
    float mindistance,
    float maxdistance
);
```

## Parameters

*position*

pointer to a vector in 3D space where the reverb is centred

*mindistance*

radius within which the reverb has full effect

*maxdistance*

radius outside of which the reverb has zero effect

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

The 3D reverb object is a sphere having 3D attributes (position, minimum distance, maximum distance) and reverb properties.

The properties and 3D attributes of all reverb objects collectively determine, along with the listener's position, the settings of and input gains into a single 3D reverb DSP.

Please note that this only applies to software channels. When the listener is within the sphere of effect of one or more 3d reverbs, the listener's 3D reverb properties are a weighted combination of such 3d reverbs. When the listener is outside all of the reverbs, the 3D reverb setting is set to the default ambient reverb setting.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [\\_EventReverb::get3DAttributes](#)

Version 4.12.03 Built on Feb 18, 2008

# EventReverb::setActive

Disables or enables a reverb object so that it does or does not contribute to the 3d scene.?

## Syntax

```
FMOD_RESULT eventReverb::setActive (
    bool active
);
```

## Parameters

*active*

true = active, false = not active. Default = true.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventReverb::getActive](#)
- [EventSystem::createReverb](#)

# EventReverb::setProperties

Sets the reverb properties for this event 3d reverb object.?

## Syntax

```
FMOD_RESULT EventReverb::setProperties (
    const FMOD_REVERB_PROPERTIES * props
);
```

## Parameters

*props*

Pointer to a [FMOD\\_REVERB\\_PROPERTIES](#) structure definition.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

[EventSystem::getReverbPreset](#) and [EventSystem::getReverbPresetByIndex](#) can be used to retrieve sound designer defined presets, or it can be set programmatically.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventReverb::getProperties](#)
- [FMOD\\_REVERB\\_PROPERTIES](#)
- [EventSystem::getReverbPreset](#)
- [EventSystem::getReverbPresetByIndex](#)
- [EventSystem::createReverb](#)

# EventReverb::setUserData

Sets a user value that the EventReverb object will store internally. Can be retrieved with [EventReverb::getUserData](#).

## Syntax

```
FMOD_RESULT eventReverb::setUserData (
    void * userData
);
```

## Parameters

*userdata*

Address of user data that the user wishes stored within the EventReverb object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

This function is primarily used in case the user wishes to 'attach' data to an FMOD object.

It can be useful if an FMOD callback passes an object of this type as a parameter, and the user does not know which object it is (if many of these types of objects exist). Using [EventReverb::getUserData](#) would help in the identification of the object.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventReverb::getUserData](#)

# Functions

[EventSystem\\_Create](#)

# EventSystem\_Create

Factory function to create an EventSystem object. This must be called to create an FMOD System object before you can do anything else.

?Use this function to create 1, or multiple instances of FMOD System objects.?

## Syntax

```
FMOD_RESULT FModSystem_Create (
    FModSystem ** pFModSystem
);
```

## Parameters

*eventsystem*

Address of a pointer that receives the new FMOD EventSystem object.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Use [EventSystem::release](#) to free an eventsystem object.

It is generally recommended to only create one system object. Creating more than one can lead to excess cpu usage as it will spawn extra software mixer threads.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, XBox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventSystem::release](#)



# Callbacks

[FMOD\\_EVENT\\_CALLBACK](#)

# FMOD\_EVENT\_CALLBACK

Event callback that is called when one of the events listed in [FMOD\\_EVENT\\_CALLBACKTYPE](#) occurs.?

## Syntax

```
FMOD_RESULT FMOD_EVENT_CALLBACK (
    FMOD_EVENT * event,
    FMOD_EVENT_CALLBACK_TYPE type,
    void * param1,
    void * param2,
    void * userdata
);
```

## Parameters

*event*

Event handle in question.

*type*

Type of callback being issued. see [FMOD\\_EVENT\\_CALLBACKTYPE](#) for the different reasons FMOD will generate a callback for an event.

*param1*

Parameter 1 for the event callback. See remarks for different uses of param1.

*param2*

Parameter 2 for the event callback. See remarks for different uses of param2.

*userdata*

User specified value set when calling [Event::setCallback](#).

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

C++ Users. Cast **FMOD\_EVENT \*** to **FMOD::Event \*** inside the callback and use as normal.

To avoid needing to process all messages simply switch on the messages you are interested in.

When the event callback is called. 'param1' and 'param2' mean different things depending on the type of callback.

Here the contents of param1 and param2 are listed.

The parameters are void \*, but should be cast to the listed C type to get the correct value.

- [FMOD\\_EVENT\\_CALLBACKTYPE\\_SYNCPOINT](#) - param1 = (char \*) name of sync point. param2 = (unsigned int) PCM offset of sync point.
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_START](#) - param1 = (char \*) name of sound definition being started. param2 = (int) index of wave being started inside sound definition (ie for multi wave sound definitions).
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_END](#) - param1 = (char \*) name of sound definition being stopped. param2 = (int) index of wave being started inside sound definition (ie for multi wave sound definitions).
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_STOLEN](#) - param1 = 0. param2 = 0. If the callback function returns [FMOD\\_ERR\\_EVENT\\_FAILED](#), the event will **not** be stolen, and the returned value will be passed back as the return value of the getEventXXX call that triggered the steal attempt.
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_EVENTFINISHED](#) - param1 = 0. param2 = 0.
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_NET\\_MODIFIED](#) - param1 = ([FMOD\\_EVENT\\_PROPERTY](#)) which property was modified. param2 = (float) the new property value.
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_CREATE](#) - param1 = (char \*) name of sound definition. param2 [in] = (int \*) pointer to index of sound definition entry. param2 [out] = (FMOD::Sound \*\*) pointer to a valid lower level API FMOD Sound handle.
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_RELEASE](#) - param1 = (char \*) name of sound definition. param2 = (FMOD::Sound \*) the FMOD sound handle that was previously created in [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_CREATE](#).
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_INFO](#) - param1 = (char \*) name of sound definition. param2 = (FMOD::Sound \*) the FMOD sound handle that FMOD will use for this sound definition.
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_EVENTSTARTED](#) - param1 = 0. param2 = 0.
- [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_SELECTINDEX](#) - param1 = (char \*) name of sound definition. param2 [in] = (int \*) pointer to number of entries in this sound definition. \*param2 [out] = (int) index of sound definition entry to select.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [FMOD\\_EVENT\\_CALLBACKTYPE](#)
- [FMOD\\_EVENT\\_PROPERTY](#)
- [Event::setCallback](#)

# Structures

[FMOD\\_EVENT\\_INFO](#)

[FMOD\\_EVENT\\_LOADINFO](#)

[FMOD\\_EVENT\\_SYSTEMINFO](#)

[FMOD\\_EVENT\\_WAVEBANKINFO](#)

# FMOD\_EVENT\_INFO

Structure containing extended information about an event.?

## Structure

```
typedef struct {
    int memoryused;
    int positionms;
    int lengthms;
    int channelsplaying;
    int instancesactive;
    char ** wavebanknames;
    unsigned int projectid;
    unsigned int systemid;
    float audibility;
    int numisamples;
    FMOD_EVENT ** instances;
} FMOD_EVENT_INFO;
```

## Members

*memoryused*

[out] Amount of memory (in bytes) used by this event. DISABLED. Not working currently until further notice. Use FMOD::Memory\_GetStats instead.

*positionms*

[out] Time passed in playback of this event instance in milliseconds.

*lengthms*

[out] Length in milliseconds of this event. Note: lengthms will be -1 if the length of the event can't be determined i.e. if it has looping sounds.

*channelsplaying*

[out] Number of channels currently playing in this event instance.

*instancesactive*

[out] Number of event instances currently in use.

*wavebanknames*

[out] An array containing the names of all wave banks that are referenced by this event.

*projectid*

[out] The runtime 'EventProject' wide unique identifier for this event.

*systemid*

[out] The runtime 'EventSystem' wide unique identifier for this event. This is calculated when single or multiple projects are loaded.

*audibility*

[out] current audibility of event.

*numinstances*

[in/out] On entry, maximum number of entries in instances array. On exit, actual number of entries in instances array, or if instances is null, then it is just the number of instances of this event. Optional.

*instances*

[in/out] Pointer to an array that will be filled with the current reference-counted event handles of all instances of this event. Optional. Specify 0 if not needed. Must be used in conjunction with numinstances. Note: Due to reference counting, the event instance handles returned here may be different between subsequent calls to this function. If you use these event handles, make sure your code is prepared for them to be invalid!

## Remarks

This structure is optional! Specify 0 or NULL in [Event::getInfo](#) if you don't need it!

This structure has members that need to be initialized before [Event::getInfo](#) is called. Always initialize this structure before calling [Event::getInfo](#)!

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [Event::getInfo](#)

# FMOD\_EVENT\_LOADINFO

Use this structure with [EventSystem::load](#) when more control is needed over loading.?

## Structure

```
typedef struct {
    unsigned int size;
    char * encryptionkey;
    float sounddefentrylimit;
    unsigned int loadfrommemory_length;
} FMOD_EVENT_LOADINFO;
```

## Members

*size*

[in] Size of this structure. This is used so the structure can be expanded in the future and still work on older versions of FMOD Ex.

*encryptionkey*

[in] Optional. Specify 0 to ignore. Key, or 'password' to decrypt a bank. A sound designer may have encrypted the audio data to protect their sound data from 'rippers'.

*sounddefentrylimit*

[in] Optional. Specify 0 to ignore. A value between 0 -> 1 that is multiplied with the number of sound definition entries in each sound definition in the project being loaded in order to programmatically reduce the number of sound definition entries used at runtime.

*loadfrommemory\_length*

[in] Optional. Specify 0 to ignore. Length of memory buffer pointed to by name\_or\_data parameter passed to [EventSystem::load](#). If this field is non-zero then the name\_or\_data parameter passed to [EventSystem::load](#) will be interpreted as a pointer to a memory buffer containing the .fev data to load. If this field is zero the name\_or\_data parameter is interpreted as the filename of the .fev file to load.

## Remarks

This structure is optional! Specify 0 or NULL in [EventSystem::load](#) if you don't need it!

Members marked with [in] mean the user sets the value before passing it to the function.

Members marked with [out] mean FMOD sets the value to be used after the function exits.

Use sounddefentrylimit to limit the number of sound definition entries - and therefore the amount of wave data - loaded for each sound definition. This feature allows the programmer to implement a "low detail" setting at runtime without needing a separate "low detail" set of assets.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [\\_EventSystem::load](#)

Version 4.12.03 Built on Feb 18, 2008



# FMOD\_EVENT\_SYSTEMINFO

Structure containing realtime information about an event system.?

## Structure

```
typedef struct {
    int    numevents;
    int    eventmemory;
    int    numinstances;
    int    instancememory;
    int    dspmemory;
    int    numwavebanks;
    FMOD_EVENT_WAVE_BANK_INFO * wavebanks;
    int    numplaygroups;
    FMOD_EVENT ** playgroups;
} FMOD_EVENT_SYSTEMINFO;
```

## Members

### *numevents*

[out] Total number of events in all event groups in this event system.

### *eventmemory*

[out] Amount of memory (in bytes) used by event hierarchy classes. DISABLED. Not working currently until further notice. Use FMOD::Memory\_GetStats instead.

### *numinstances*

[out] Total number of event instances in all event groups in this event system.

### *instancememory*

[out] Amount of memory (in bytes) used by all event instances in this event system. DISABLED. Not working currently until further notice. Use FMOD::Memory\_GetStats instead.

### *dspmemory*

[out] Amount of memory (in bytes) used by event dsp networks. DISABLED. Not working currently until further notice. Use FMOD::Memory\_GetStats instead.

### *numwavebanks*

[out] Number of wave banks known to this event system.

### *wavebankinfo*

[out] Array of detailed information on each wave bank.

*numplayingevents*

[in/out] On entry, maximum number of entries in playingevents array. On exit, actual number of entries in playingevents array, or if playingevents is null, then it is just the number of currently playing events. Optional.

*playingevents*

[in/out] Pointer to an array that will be filled with the event handles of all playing events. Optional. Specify 0 if not needed. Must be used in conjunction with numplayingevents.

## Remarks

On entry, numplayingevents should be set to the number of elements in the playingevents array. If the actual number of playing events is greater than numplayingevents then the playingevents array will be filled with numplayingevents entries and numplayingevents will be set to the actual number of playing events on exit. In short, if numplayingevents on exit > numplayingevents on entry then the playingevents array wasn't large enough and some events were unable to be added to the array.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventSystem::getInfo](#)
- [FMOD\\_EVENT\\_WAVEBANKINFO](#)

# FMOD\_EVENT\_WAVEBANKINFO

Structure containing realtime information about a wavebank.?

## Structure

```
typedef struct {
    char * name;
    int stream_refcnt;
    int sample_refcnt;
    int numstreams;
    int maxstreams;
    int streams_inuse;
    unsigned int streammemory;
    unsigned int samplememory;
} FMOD_EVENT_WAVEBANKINFO;
```

## Members

*name*

[out] Name of this wave bank.

*streamrefcnt*

[out] Number of stream references to this wave bank made by events in this event system.

*samplerefcnt*

[out] Number of sample references to this wave bank made by events in this event system.

*numstreams*

[out] Number of times this wave bank has been opened for streaming.

*maxstreams*

[out] Maximum number of times this wave bank will be opened for streaming.

*streams\_inuse*

[out] Number of streams currently in use.

*streammemory*

[out] Amount of memory (in bytes) used by streams.

*samplememory*

[out] Amount of memory (in bytes) used by samples.

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [\\_EventSystem::getInfo](#)
- [EMOD\\_EVENT\\_SYSTEMINFO](#)

# Defines

[EVENT\\_INITFLAGS](#)

[EVENT\\_MODE](#)

[EVENT\\_STATE](#)

# FMOD\_EVENT\_INITFLAGS

Initialization flags. Use them with [EventSystem::init](#) in the eventflags parameter to change various behaviour.?

## Definition

```
# d e f n e F M O D _ E V E N T _ I N I T _ N O R M A L 0x00000000
# d e f n e F M O D _ E V E N T _ I N I T _ U S E _ R A S S E _ M A N A G E R 0x00000001
# d e f n e F M O D _ E V E N T _ I N I T _ F A I L _ O N _ M A X _ S T R E A M S 0x00000002
# d e f n e F M O D _ E V E N T _ I N I T _ D O N T _ U S E _ N A M E S 0x00000004
# d e f n e F M O D _ E V E N T _ I N I T _ U P P E R _ C A S E _ F I L E _ N A M E S 0x00000008
# d e f n e F M O D _ E V E N T _ I N I T _ S E A R C H _ P L U G I N S 0x00000010
```

## Values

*FMOD\_EVENT\_INIT\_NORMAL*

All platforms - Initialize normally

*FMOD\_EVENT\_INIT\_USER\_ASSETMANAGER*

All platforms - All wave data loading/freeing will be referred back to the user through the event callback

*FMOD\_EVENT\_INIT\_FAIL\_ON\_MAXSTREAMS*

All platforms - Events will fail if "Max streams" was reached when playing streamed banks, instead of going virtual.

*FMOD\_EVENT\_INIT\_DONTUSENAMES*

All platforms - All event/eventgroup/eventparameter/eventcategory/eventreverb names will be discarded on load. Use `getXXXByIndex` to access them. This may potentially save a lot of memory at runtime.

*FMOD\_EVENT\_INIT\_UPPERCASE\_FILENAMES*

All platforms - All FSB filenames will be translated to upper case before being used.

*FMOD\_EVENT\_INIT\_SEARCH\_PLUGINS*

All platforms - Search the current directory for dsp/codec plugins on [EventSystem::init](#).

## Platforms Supported

Win32, Win64, Linux, Linux64, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3

## See Also

- [EventSystem::init](#)

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# FMOD\_EVENT\_MODE

Event data loading bitfields. Bitwise OR them together for controlling how event data is loaded.?

## Definition

```
# d f a F M O _ D E V E N T _ E R R O R _ O N _ D I S K A C C E S S 0x00000002
# d f a F M O _ D E V E N T _ N O N B L O C K I N G 0x00000004
# d f a F M O _ D E V E N T _ N O N C A C H E D 0x00000001
# d f a F M O _ D E V E N T _ D E F A U L T 0x00000000
```

## Values

*FMOD\_EVENT\_DEFAULT*

FMOD\_EVENT\_DEFAULT specifies default loading behaviour i.e. event data for the whole group is NOT cached and the function that initiated the loading process will block until loading is complete.

*FMOD\_EVENT\_NONBLOCKING*

For loading event data asynchronously. FMOD will use a thread to load the data. Use `Event::getState` to find out when loading is complete.

*FMOD\_EVENT\_ERROR\_ON\_DISKACCESS*

For [EventGroup::getEvent](#) / [EventGroup::getEventByIndex](#). If [EventGroup::loadEventData](#) has accidentally been forgotten this flag will return an FMOD\_ERR\_FILE\_UNWANTED if the `getEvent` function tries to load data.

*FMOD\_EVENT\_INFOONLY*

For [EventGroup::getEvent](#) / [EventGroup::getEventByIndex](#). Don't allocate instances or load data, just get a handle to allow user to get information from the event.

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::loadEventData](#)
- [EventGroup::getEvent](#)
- [EventGroup::getEventByIndex](#)





# FMOD\_EVENT\_STATE

These values describe what state an event is in. The flags below can be combined to set multiple states at once. Use bitwise AND operations to test for these. An example of a combined flag set would be FMOD\_EVENT\_STATE\_READY | FMOD\_EVENT\_STATE\_PLAYING.

## Definition

```
#define FMOD_EVENT_STATE_READY 0x00000001
#define FMOD_EVENT_STATE_LOADING 0x00000002
#define FMOD_EVENT_STATE_ERROR 0x00000004
#define FMOD_EVENT_STATE_PLAYING 0x00000008
#define FMOD_EVENT_STATE_CHANNELSACTIVE 0x00000010
#define FMOD_EVENT_STATE_INFOONLY 0x00000020
#define FMOD_EVENT_STATE_STARVING 0x00000040
```

## Values

*FMOD\_EVENT\_STATE\_READY*

Event is ready to play.

*FMOD\_EVENT\_STATE\_LOADING*

Loading in progress.

*FMOD\_EVENT\_STATE\_ERROR*

Failed to open - file not found, out of memory etc. See return value of [Event::getState](#) for what happened.

*FMOD\_EVENT\_STATE\_PLAYING*

Event has been started. This will still be true even if there are no sounds active. Event::stop must be called or the event must stop itself using a 'one shot and stop event' parameter mode.

*FMOD\_EVENT\_STATE\_CHANNELSACTIVE*

Event has active voices. Use this if you want to detect if sounds are playing in the event or not.

*FMOD\_EVENT\_STATE\_INFOONLY*

Event was loaded with the FMOD\_EVENT\_INFOONLY flag.

*FMOD\_EVENT\_STATE\_STARVING*

Event is streaming but not being fed data in time, so may be stuttering.

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

#### See Also

- [Event::getState](#)
- [FMOD\\_EVENT\\_MODE](#)

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# Enumerations

[FMOD\\_EVENT\\_CALLBACKTYPE](#)

[FMOD\\_EVENT\\_PITCHUNITS](#)

[FMOD\\_EVENT\\_PROPERTY](#)

[FMOD\\_EVENT\\_RESOURCE](#)

# FMOD\_EVENT\_CALLBACKTYPE

These callback types are used with [FMOD\\_EVENT\\_CALLBACK](#).

## Enumeration

```
typedef enum {
    FMOD_EVENT_CALLBACKTYPE_SYNCPOINT,
    FMOD_EVENT_CALLBACKTYPE_SOUNDDEF_START,
    FMOD_EVENT_CALLBACKTYPE_SOUNDDEF_END,
    FMOD_EVENT_CALLBACKTYPE_SOUND,
    FMOD_EVENT_CALLBACKTYPE_EVENTINSTANCE,
    FMOD_EVENT_CALLBACKTYPE_EVENTMODIFIED,
    FMOD_EVENT_CALLBACKTYPE_SOUNDDEF_CREATE,
    FMOD_EVENT_CALLBACKTYPE_SOUNDDEF_RELEASE,
    FMOD_EVENT_CALLBACKTYPE_SOUNDDEF_FINISHED,
    FMOD_EVENT_CALLBACKTYPE_EVENTINSTANCES,
    FMOD_EVENT_CALLBACKTYPE_SOUNDDEF_DESTROY
}
```

## Values

*FMOD\_EVENT\_CALLBACKTYPE\_SYNCPOINT*

Called when a syncpoint is encountered. Can be from wav file markers.

*FMOD\_EVENT\_CALLBACKTYPE\_SOUNDDEF\_START*

Called when a sound definition inside an event is triggered.

*FMOD\_EVENT\_CALLBACKTYPE\_SOUNDDEF\_END*

Called when a sound definition inside an event ends or is stopped.

*FMOD\_EVENT\_CALLBACKTYPE\_STOLEN*

Called when a `getEventXXX` call steals a playing event instance.

*FMOD\_EVENT\_CALLBACKTYPE\_EVENTFINISHED*

Called when an event is stopped for any reason.

*FMOD\_EVENT\_CALLBACKTYPE\_NET\_MODIFIED*

Called when a property of the event has been modified by a network-connected host.

*FMOD\_EVENT\_CALLBACKTYPE\_SOUNDDEF\_CREATE*

Called when a programmer sound definition entry is loaded.

*FMOD\_EVENT\_CALLBACKTYPE\_SOUNDDEF\_RELEASE*

Called when a programmer sound definition entry is unloaded.

*FMOD\_EVENT\_CALLBACKTYPE\_SOUNDDEF\_INFO*

Called when a sound definition entry is loaded.

*FMOD\_EVENT\_CALLBACKTYPE\_EVENTSTARTED*

Called when an event is started.

*FMOD\_EVENT\_CALLBACKTYPE\_SOUNDDEF\_SELECTINDEX*

Called when a sound definition entry needs to be chosen from a "ProgrammerSelected" sound definition.

## Remarks

**Note!** Currently the user must call [EventSystem:update](#) for these callbacks to trigger!

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_SYNCPOINT](#) callback is generated from 'markers' embedded in .wav files. These can be created by placing 'markers' in the original source wavs using a tool such as Sound Forge or Cooledit.

The wavs are then compiled into .FSB files when compiling the audio data using the FMOD designer tool.

Callbacks will be automatically generated at the correct place in the timeline when these markers are encountered which makes it useful for synchronization, lip syncing etc.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_START](#) callback is generated each time a sound definition is played in an event.

This happens every time a sound definition starts due to the event parameter entering the region specified in the layer created by the sound designer..

This also happens when sounds are randomly respawned using the random respawn feature in the sound definition properties in FMOD designer.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_END](#) callback is generated when a one-shot sound definition inside an event ends, or when a looping sound definition stops due to the event parameter leaving the region specified in the layer created by the sound designer.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_STOLEN](#) callback is generated when a `getEventXXX` call needs to steal a playing event because the event's "Max playbacks" has been exceeded. This callback is called before the event is stolen and before the event is stopped. An [FMOD\\_EVENT\\_CALLBACKTYPE\\_EVENTFINISHED](#) callback will be generated when the stolen event is stopped i.e. **after** the [FMOD\\_EVENT\\_CALLBACKTYPE\\_STOLEN](#). If the callback function returns [FMOD\\_ERR\\_EVENT\\_FAILED](#), the event will **not** be stolen, and the returned value will be passed back as the return value of the `getEventXXX` call that triggered the steal attempt.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_EVENTFINISHED](#) callback is generated whenever an event is stopped for any reason including when the user calls `Event::stop()`.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_NET\\_MODIFIED](#) callback is generated when someone has connected to your running application with FMOD Designer and changed a property within this event, for example volume or pitch.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_CREATE](#) callback is generated when a "programmer"

sound needs to be loaded.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_RELEASE](#) callback is generated when a "programmer" sound needs to be unloaded.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_INFO](#) callback is generated when a sound definition is loaded. It can be used to find information about the specific sound that will be played.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_EVENTSTARTED](#) callback is generated whenever an event is started. This callback will be called before any sounds in the event have begun to play.

An [FMOD\\_EVENT\\_CALLBACKTYPE\\_SOUNDDEF\\_SELECTINDEX](#) callback is generated when a sound definition entry needs to be chosen from a "ProgrammerSelected" sound definition.

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii, Wii

## See Also

- [Event::setCallback](#)
- [FMOD\\_EVENT\\_CALLBACK](#)
- [EventSystem::update](#)

# FMOD\_EVENT\_PITCHUNITS

Pitch units for [Event::setPitch](#) and [EventCategory::setPitch](#).

## Enumeration

```
typedef enum {  
    FMOD_EVENT_PITCHUNITS_RAW,  
    FMOD_EVENT_PITCHUNITS_OCTAVES,  
    FMOD_EVENT_PITCHUNITS_SEMITONES,  
    FMOD_EVENT_PITCHUNITS_TONES  
} FMOD_EVENT_PITCHUNITS;
```

## Values

*FMOD\_EVENT\_PITCHUNITS\_RAW*

Pitch is specified in raw underlying units.

*FMOD\_EVENT\_PITCHUNITS\_OCTAVES*

Pitch is specified in units of octaves.

*FMOD\_EVENT\_PITCHUNITS\_SEMITONES*

Pitch is specified in units of semitones.

*FMOD\_EVENT\_PITCHUNITS\_TONES*

Pitch is specified in units of tones.

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [Event::setPitch](#)
- [EventCategory::setPitch](#)



# FMOD\_EVENT\_PROPERTY

Property indices for [Event::getPropertyByIndex](#)?

## Enumeration

```
typedef enum {
    FMOD_EVENTPROPERTY_NAME,
    FMOD_EVENTPROPERTY_VOLUME,
    FMOD_EVENTPROPERTY_VOLUME_ADMISSION,
    FMOD_EVENTPROPERTY_PITCH,
    FMOD_EVENTPROPERTY_PITCH_HOCTES,
    FMOD_EVENTPROPERTY_PITCH_SEMIDBS,
    FMOD_EVENTPROPERTY_PITCH_DBFS,
    FMOD_EVENTPROPERTY_PITCH_ADMISSION,
    FMOD_EVENTPROPERTY_PITCH_ADMISSION_NOCUES,
    FMOD_EVENTPROPERTY_PITCH_ADMISSION_NSEMI_DBFS,
    FMOD_EVENTPROPERTY_PITCH_ADMISSION_N_DBFS,
    FMOD_EVENTPROPERTY_PROBET,
    FMOD_EVENTPROPERTY_MAXPAYBACKS,
    FMOD_EVENTPROPERTY_MAXPAYBACKS_BAYOR,
    FMOD_EVENTPROPERTY_MOSH,
    FMOD_EVENTPROPERTY_3D_BLEND,
    FMOD_EVENTPROPERTY_3D_MINDSANE,
    FMOD_EVENTPROPERTY_3D_MAXDMSANE,
    FMOD_EVENTPROPERTY_3D_DSITON,
    FMOD_EVENTPROPERTY_3D_COINTEGRATE,
    FMOD_EVENTPROPERTY_3D_COINTEGRATE,
    FMOD_EVENTPROPERTY_3D_COINTEGRATE_VOLUME,
    FMOD_EVENTPROPERTY_3D_DDPESCAE,
    FMOD_EVENTPROPERTY_3D_SCAKEPREAD,
    FMOD_EVENTPROPERTY_3D_DANEWL,
    FMOD_EVENTPROPERTY_SCAKE_RL,
    FMOD_EVENTPROPERTY_SCAKE_RC,
    FMOD_EVENTPROPERTY_SCAKE_RR,
    FMOD_EVENTPROPERTY_SCAKE_RS,
    FMOD_EVENTPROPERTY_SCAKE_RB,
    FMOD_EVENTPROPERTY_SCAKE_RLR,
    FMOD_EVENTPROPERTY_SCAKE_RRR,
    FMOD_EVENTPROPERTY_SCAKE_RLE,
    FMOD_EVENTPROPERTY_EMBEDTEWL,
    FMOD_EVENTPROPERTY_ONSET,
    FMOD_EVENTPROPERTY_AFIN,
    FMOD_EVENTPROPERTY_AHOUT,
    FMOD_EVENTPROPERTY_EMBEDTEWL,
    FMOD_EVENTPROPERTY_TMEOFFSET,
    FMOD_EVENTPROPERTY_SAWNNESIT,
    FMOD_EVENTPROPERTY_SAWNNESIT_ADMISSION,
    FMOD_EVENTPROPERTY_WII_CONTROLLER_SCAKE_R,
    FMOD_EVENTPROPERTY_3D_DS_ADMISSION,
    FMOD_EVENTPROPERTY_USE_RASE
}
```

## Values

*FMOD\_EVENTPROPERTY\_NAME*

Type : char \* - Name of event.

*FMOD\_EVENTPROPERTY\_VOLUME*

Type : float - Relative volume of event.

*FMOD\_EVENTPROPERTY\_VOLUMERANDOMIZATION*

Type : float - Random deviation in volume of event.

*FMOD\_EVENTPROPERTY\_PITCH*

Type : float - Relative pitch of event in raw underlying units.

*FMOD\_EVENTPROPERTY\_PITCH\_OCTAVES*

Type : float - Relative pitch of event in octaves.

*FMOD\_EVENTPROPERTY\_PITCH\_SEMITONES*

Type : float - Relative pitch of event in semitones.

*FMOD\_EVENTPROPERTY\_PITCH\_TONES*

Type : float - Relative pitch of event in tones.

*FMOD\_EVENTPROPERTY\_PITCHRANDOMIZATION*

Type : float - Random deviation in pitch of event in raw underlying units.

*FMOD\_EVENTPROPERTY\_PITCHRANDOMIZATION\_OCTAVES*

Type : float - Random deviation in pitch of event in octaves.

*FMOD\_EVENTPROPERTY\_PITCHRANDOMIZATION\_SEMITONES*

Type : float - Random deviation in pitch of event in semitones.

*FMOD\_EVENTPROPERTY\_PITCHRANDOMIZATION\_TONES*

Type : float - Random deviation in pitch of event in tones.

*FMOD\_EVENTPROPERTY\_PRIORITY*

Type : int - Playback priority of event.

*FMOD\_EVENTPROPERTY\_MAX\_PLAYBACKS*

Type : int - Maximum simultaneous playbacks of event.

*FMOD\_EVENTPROPERTY\_MAX\_PLAYBACKS\_BEHAVIOR*

Type : int - 1 = steal oldest, 2 = steal newest, 3 = steal quietest, 4 = just fail, 5 = just fail if quietest.

## *FMOD\_EVENTPROPERTY\_MODE*

Type : [FMOD\\_MODE](#) - Either [FMOD\\_3D](#) or [FMOD\\_2D](#).

## *FMOD\_EVENTPROPERTY\_3D\_ROLLOFF*

Type : [FMOD\\_MODE](#) - Either [FMOD\\_3D\\_LOGROLLOFF](#), [FMOD\\_3D\\_LINEARROLLOFF](#), or none for custom rolloff.

## *FMOD\_EVENTPROPERTY\_3D\_MINDISTANCE*

Type : float - Minimum 3d distance of event.

## *FMOD\_EVENTPROPERTY\_3D\_MAXDISTANCE*

Type : float - Maximum 3d distance of event. Means different things depending on [EVENTPROPERTY\\_3D\\_ROLLOFF](#). If event has custom rolloff, setting [FMOD\\_EVENTPROPERTY\\_3D\\_MAXDISTANCE](#) will scale the range of all distance parameters in this event e.g. set this property to 2.0 to double the range of all distance parameters, set it to 0.5 to halve the range of all distance parameters.

## *FMOD\_EVENTPROPERTY\_3D\_POSITION*

Type : [FMOD\\_MODE](#) - Either [FMOD\\_3D\\_HEADRELATIVE](#) or [FMOD\\_3D\\_WORLDRELATIVE](#).

## *FMOD\_EVENTPROPERTY\_3D\_CONEINSIDEANGLE*

Type : float - Event cone inside angle. 0 to 360.

## *FMOD\_EVENTPROPERTY\_3D\_CONEOUTSIDEANGLE*

Type : float - Event cone outside angle. 0 to 360.

## *FMOD\_EVENTPROPERTY\_3D\_CONEOUTSIDEVOLUME*

Type : float - Event cone outside volume. 0 to 1.0.

## *FMOD\_EVENTPROPERTY\_3D\_DOPPLERSCALE*

Type : float - Doppler scale where 0 = no doppler, 1.0 = normal doppler, 2.0 = double doppler etc.

## *FMOD\_EVENTPROPERTY\_3D\_SPEAKERSPREAD*

Type : float - Angle of spread for stereo/multichannel source. 0 to 360.

## *FMOD\_EVENTPROPERTY\_3D\_PANLEVEL*

Type : float - 0 = sound pans according to speaker levels, 1 = sound pans according to 3D position.

## *FMOD\_EVENTPROPERTY\_SPEAKER\_L*

Type : float - 2D event volume for front left speaker.

## *FMOD\_EVENTPROPERTY\_SPEAKER\_C*

Type : float - 2D event volume for front left speaker.

*FMOD\_EVENTPROPERTY\_SPEAKER\_R*

Type : float - 2D event volume for front left speaker.

*FMOD\_EVENTPROPERTY\_SPEAKER\_LS*

Type : float - 2D event volume for front left speaker.

*FMOD\_EVENTPROPERTY\_SPEAKER\_RS*

Type : float - 2D event volume for front left speaker.

*FMOD\_EVENTPROPERTY\_SPEAKER\_LR*

Type : float - 2D event volume for front left speaker.

*FMOD\_EVENTPROPERTY\_SPEAKER\_RR*

Type : float - 2D event volume for front left speaker.

*FMOD\_EVENTPROPERTY\_SPEAKER\_LFE*

Type : float - 2D event volume for front left speaker.

*FMOD\_EVENTPROPERTY\_REVERBWETLEVEL*

Type : float - Reverb gain for this event where 0 = full reverb, -60 = no reverb.

*FMOD\_EVENTPROPERTY\_ONESHOT*

Type : int - Oneshot event - stops when no channels playing

*FMOD\_EVENTPROPERTY\_FADEIN*

Type : int - Time in milliseconds over which to fade this event in when programmer starts it. 0 = no fade in.

*FMOD\_EVENTPROPERTY\_FADEOUT*

Type : int - Time in milliseconds over which to fade this event out when programmer stops it. 0 = no fade out.

*FMOD\_EVENTPROPERTY\_REVERBDRYLEVEL*

Type : float - Dry reverb gain for this event where 0 = full dry, -60 = no dry.

*FMOD\_EVENTPROPERTY\_TIMEOFFSET*

Type : float - Time offset of sound start in seconds (0 to 60.0f)

*FMOD\_EVENTPROPERTY\_SPAWNINTENSITY*

Type : float - Multiplier for spawn frequency of all sounds in this event.

*FMOD\_EVENTPROPERTY\_SPAWNINTENSITY\_RANDOMIZATION*

Type : float - Random deviation in spawn intensity of event.

#### *FMOD\_EVENTPROPERTY\_WII\_CONTROLLERSPEAKER*

Type : int - Wii only. Use 0 to 3 to specify a Wii controller speaker to play this event on, -1 to play on normal Wii speakers.

#### *FMOD\_EVENTPROPERTY\_3D\_POSRANDOMIZATION*

Type : unsigned int - Radius of random deviation in the 3D position of event.

#### *FMOD\_EVENTPROPERTY\_USER\_BASE*

User created events start from here onwards.

### **Platforms Supported**

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

### **See Also**

- [Event::getPropertyByIndex](#)

# FMOD\_EVENT\_RESOURCE

Flags to pass to [EventGroup::loadEventData](#) to determine what to load at the time of calling.?

## Enumeration

```
typedef enum {  
    FMOD_EVENT_RESOURCE_STREAMS_AND_SAMPLES ,  
    FMOD_EVENT_RESOURCE_STREAMS ,  
    FMOD_EVENT_RESOURCE_SAMPLES  
} FMOD_EVENT_RESOURCE;
```

## Values

*FMOD\_EVENT\_RESOURCE\_STREAMS\_AND\_SAMPLES*

Open all streams and load all banks into memory, under this group (recursive)

*FMOD\_EVENT\_RESOURCE\_STREAMS*

Open all streams under this group (recursive). No samples are loaded.

*FMOD\_EVENT\_RESOURCE\_SAMPLES*

Load all banks into memory, under this group (recursive). No streams are opened.

## Platforms Supported

Win32, Win64, Linux, Macintosh, Xbox, Xbox360, PlayStation 2, GameCube, PlayStation Portable, PlayStation 3, Wii

## See Also

- [EventGroup::loadEventData](#)

# C++ Reference

[Functions](#)

# NetEventSystem\_GetVersion

Get the NetEventSystem version number.?

## Syntax

```
FMOD_RESULT NetEventSystem_GetVersion(  
    unsigned int * version  
);
```

## Parameters

*version*

A pointer to an integer to receive the version number

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Platforms Supported

## See Also

- [\\_NetEventSystem\\_Init](#)



# NetEventSystem\_Init

This function initializes the NetEventSystem and prepares it to accept incoming connections. ?NOTE: This function must be called before any other NetEventSystem functions.?

## Syntax

```
FMOD_RESULT NetEventSystem_Init(  
    EventSystem *eventsystem,  
    unsigned short port  
);
```

## Parameters

*eventsystem*

A pointer to a user-created EventSystem object.

*port*

The TCP port that the NetEventSystem will use to accept incoming connections. 0 = use default port which is 17997.

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Specify 0 for the port unless you have a good reason not to. Make sure that whatever port you specify is not blocked.

## Platforms Supported

## See Also

- [NetEventSystem\\_Update](#)
- [NetEventSystem\\_Shutdown](#)
- [NetEventSystem\\_GetVersion](#)



# NetEventSystem\_Shutdown

Shut down the NetEventSystem.?

## Syntax

```
FMOD_RESULT NetEventSystem_Shutdown();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

Call this function after you call EventSystem:release.

## Platforms Supported

## See Also

- [\\_NetEventSystem\\_Init](#)

# NetEventSystem\_Update

Update the NetEventSystem.

## Syntax

```
FMOD_RESULT NetEventSystem_Update();
```

## Parameters

## Return Values

If the function succeeds then the return value is [FMOD\\_OK](#).

If the function fails then the return value will be one of the values defined in the [FMOD\\_RESULT](#) enumeration.

## Remarks

You must call this function once a frame just after you call EventSystem:update.

## Platforms Supported

## See Also

- [NetEventSystem\\_Init](#)
- [NetEventSystem\\_Shutdown](#)