

Photon Voice

v2.22.3

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Chapter 1

Main Page

Photon Voice 2 has three key classes:

- `Photon.Voice.Unity.VoiceConnection` (extends `Photon.Realtime.LoadBalancingClient`)
- `Photon.Voice.Unity.Recorder`
- `Photon.Voice.Unity.Speaker`

If you also use the integration with PUN 2, we added two components for ease-of-use and more convenience:

- `Photon.Voice.PUN.PhotonVoiceNetwork`
- `Photon.Voice.PUN.PhotonVoiceView`

Photon Voice 2 also comes with a WebRTC based DSP (`Photon.Voice.Unity.WebRtcAudioDsp` using `Photon.Voice.WebRTCAudioProcessor`).

Read more in the official documentation [here](#).

You can download Photon Voice 2 [here](#).

Chapter 2

NOTE:

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Chapter 3

Namespace Documentation

3.1 CSCore Namespace Reference

Classes

- class [AudioSubTypes](#)
Defines [AudioSubTypes](#) and provides methods to convert between [AudioEncoding](#)-values and [AudioSubTypes](#)-values.
- class [Extensions](#)
Provides a few basic extensions.
- interface [IAudioSource](#)
Defines the base for all audio streams.
- interface [IReadableAudioSource](#)
Defines a generic base for all readable audio streams.
- interface [IWaveSource](#)
Defines the base for all audio streams which provide raw byte data.
- interface [IWriteable](#)
Provides the [Write](#) method.
- class [WaveFormat](#)
Defines the format of waveform-audio data.
- class [WaveFormatExtensible](#)
Defines the format of waveform-audio data for formats having more than two channels or higher sample resolutions than allowed by [WaveFormat](#). Can be used to define any format that can be defined by [WaveFormat](#). For more information see [and](#) .

Enumerations

- enum [AudioEncoding](#) : short
Defines all known encoding types. Primary used in the [WaveFormat](#) class. See [WaveFormat.WaveFormatTag](#).
- enum [ChannelMask](#)
Channelmask used by [WaveFormatExtensible](#). For more information see [http://msdn.microsoft.com/en-us/library/windows/desktop/dd757714\(v=vs.85\).aspx](http://msdn.microsoft.com/en-us/library/windows/desktop/dd757714(v=vs.85).aspx)

3.1.1 Enumeration Type Documentation

3.1.1.1 AudioEncoding

enum [AudioEncoding](#) : short [strong]

Defines all known encoding types. Primary used in the [WaveFormat](#) class. See [WaveFormat.WaveFormatTag](#).

Enumerator

Unknown	WAVE_FORMAT_UNKNOWN, Microsoft Corporation
Pcm	WAVE_FORMAT_PCM Microsoft Corporation
Adpcm	WAVE_FORMAT_ADPCM Microsoft Corporation
IeeeFloat	WAVE_FORMAT_IEEE_FLOAT Microsoft Corporation
Vselp	WAVE_FORMAT_VSELP Compaq Computer Corp.
IbmCvsd	WAVE_FORMAT_IBM_CVSD IBM Corporation
ALaw	WAVE_FORMAT_ALAW Microsoft Corporation
MuLaw	WAVE_FORMAT_MULAW Microsoft Corporation
Dts	WAVE_FORMAT_DTS Microsoft Corporation
Drm	WAVE_FORMAT_DRM Microsoft Corporation
WmaVoice9	WAVE_FORMAT_WMAVOICE9
OkiAdpcm	WAVE_FORMAT_OKI_ADPCM OKI
DviAdpcm	WAVE_FORMAT_DVI_ADPCM Intel Corporation
ImaAdpcm	WAVE_FORMAT_IMA_ADPCM Intel Corporation
MediaspaceAdpcm	WAVE_FORMAT_MEDIASPACE_ADPCM Videologic
SierraAdpcm	WAVE_FORMAT_SIERRA_ADPCM Sierra Semiconductor Corp
G723Adpcm	WAVE_FORMAT_G723_ADPCM Antex Electronics Corporation
DigiStd	WAVE_FORMAT_DIGISTD DSP Solutions, Inc.
DigiFix	WAVE_FORMAT_DIGIFIX DSP Solutions, Inc.
DialogicOkiAdpcm	WAVE_FORMAT_DIALOGIC_OKI_ADPCM Dialogic Corporation
MediaVisionAdpcm	WAVE_FORMAT_MEDIAVISION_ADPCM Media Vision, Inc.
CUCodec	WAVE_FORMAT_CU_CODEC Hewlett-Packard Company
YamahaAdpcm	WAVE_FORMAT_YAMAHA_ADPCM Yamaha Corporation of America
SonarC	WAVE_FORMAT_SONARC Speech Compression
DspGroupTrueSpeech	WAVE_FORMAT_DSPGROUP_TRUESPEECH DSP Group, Inc
EchoSpeechCorporation1	WAVE_FORMAT_ECHOSC1 Echo Speech Corporation
AudioFileAf36	WAVE_FORMAT_AUDIOFILE_AF36, Virtual Music, Inc.
Aptx	WAVE_FORMAT_APTX Audio Processing Technology
AudioFileAf10	WAVE_FORMAT_AUDIOFILE_AF10, Virtual Music, Inc.
Prosody1612	WAVE_FORMAT_PROSODY_1612, Aculab plc
Lrc	WAVE_FORMAT_LRC, Merging Technologies S.A.
DolbyAc2	WAVE_FORMAT_DOLBY_AC2, Dolby Laboratories

Enumerator

Gsm610	WAVE_FORMAT_GSM610, Microsoft Corporation
MsnAudio	WAVE_FORMAT_MSNAUDIO, Microsoft Corporation
AntexAdpcm	WAVE_FORMAT_ANTEX_ADPCME, Antex Electronics Corporation
ControlResVqlpc	WAVE_FORMAT_CONTROL_RES_VQLPC, Control Resources Limited
DigiReal	WAVE_FORMAT_DIGIREAL, DSP Solutions, Inc.
DigiAdpcm	WAVE_FORMAT_DIGIADPCM, DSP Solutions, Inc.
ControlResCr10	WAVE_FORMAT_CONTROL_RES_CR10, Control Resources Limited
WAVE_FORMAT_NMS_VBXADPCM	WAVE_FORMAT_NMS_VBXADPCM
WAVE_FORMAT_CS_IMAADPCM	WAVE_FORMAT_CS_IMAADPCM
WAVE_FORMAT_ECHOSC3	WAVE_FORMAT_ECHOSC3
WAVE_FORMAT_ROCKWELL_ADPCM	WAVE_FORMAT_ROCKWELL_ADPCM
WAVE_FORMAT_ROCKWELL_DIGITALK	WAVE_FORMAT_ROCKWELL_DIGITALK
WAVE_FORMAT_XEBEC	WAVE_FORMAT_XEBEC
WAVE_FORMAT_G721_ADPCM	WAVE_FORMAT_G721_ADPCM
WAVE_FORMAT_G728_CELP	WAVE_FORMAT_G728_CELP
WAVE_FORMAT_MSG723	WAVE_FORMAT_MSG723
Mpeg	WAVE_FORMAT_MPEG, Microsoft Corporation
WAVE_FORMAT_RT24	WAVE_FORMAT_RT24
WAVE_FORMAT_PAC	WAVE_FORMAT_PAC
MpegLayer3	WAVE_FORMAT_MPEGLAYER3, ISO/MPEG Layer3 Format Tag
WAVE_FORMAT_LUCENT_G723	WAVE_FORMAT_LUCENT_G723
WAVE_FORMAT_CIRRUS	WAVE_FORMAT_CIRRUS
WAVE_FORMAT_ESPCM	WAVE_FORMAT_ESPCM
WAVE_FORMAT_VOXWARE	WAVE_FORMAT_VOXWARE
WAVE_FORMAT_CANOPUS_ATRAC	WAVE_FORMAT_CANOPUS_ATRAC
WAVE_FORMAT_G726_ADPCM	WAVE_FORMAT_G726_ADPCM
WAVE_FORMAT_G722_ADPCM	WAVE_FORMAT_G722_ADPCM
WAVE_FORMAT_DSAT_DISPLAY	WAVE_FORMAT_DSAT_DISPLAY
WAVE_FORMAT_VOXWARE_BYTE_ALIGNED	WAVE_FORMAT_VOXWARE_BYTE_ALIGNED
WAVE_FORMAT_VOXWARE_AC8	WAVE_FORMAT_VOXWARE_AC8
WAVE_FORMAT_VOXWARE_AC10	WAVE_FORMAT_VOXWARE_AC10
WAVE_FORMAT_VOXWARE_AC16	WAVE_FORMAT_VOXWARE_AC16
WAVE_FORMAT_VOXWARE_AC20	WAVE_FORMAT_VOXWARE_AC20
WAVE_FORMAT_VOXWARE_RT24	WAVE_FORMAT_VOXWARE_RT24
WAVE_FORMAT_VOXWARE_RT29	WAVE_FORMAT_VOXWARE_RT29
WAVE_FORMAT_VOXWARE_RT29HW	WAVE_FORMAT_VOXWARE_RT29HW
WAVE_FORMAT_VOXWARE_VR12	WAVE_FORMAT_VOXWARE_VR12
WAVE_FORMAT_VOXWARE_VR18	WAVE_FORMAT_VOXWARE_VR18
WAVE_FORMAT_VOXWARE_TQ40	WAVE_FORMAT_VOXWARE_TQ40
WAVE_FORMAT_SOFTSOUND	WAVE_FORMAT_SOFTSOUND
WAVE_FORMAT_VOXWARE_TQ60	WAVE_FORMAT_VOXWARE_TQ60
WAVE_FORMAT_MSRT24	WAVE_FORMAT_MSRT24
WAVE_FORMAT_G729A	WAVE_FORMAT_G729A
WAVE_FORMAT_MVI_MVI2	WAVE_FORMAT_MVI_MVI2
WAVE_FORMAT_DF_G726	WAVE_FORMAT_DF_G726

Enumerator

WAVE_FORMAT_DF_GSM610	WAVE_FORMAT_DF_GSM610
WAVE_FORMAT_ISIAUDIO	WAVE_FORMAT_ISIAUDIO
WAVE_FORMAT_ONLIVE	WAVE_FORMAT_ONLIVE
WAVE_FORMAT_SBC24	WAVE_FORMAT_SBC24
WAVE_FORMAT_DOLBY_AC3_SPDIF	WAVE_FORMAT_DOLBY_AC3_SPDIF
WAVE_FORMAT_MEDIASONIC_G723	WAVE_FORMAT_MEDIASONIC_G723
WAVE_FORMAT_PROSODY_8KBPS	WAVE_FORMAT_PROSODY_8KBPS
WAVE_FORMAT_ZYXEL_ADPCM	WAVE_FORMAT_ZYXEL_ADPCM
WAVE_FORMAT_PHILIPS_LPCBB	WAVE_FORMAT_PHILIPS_LPCBB
WAVE_FORMAT_PACKED	WAVE_FORMAT_PACKED
WAVE_FORMAT_MALDEN_PHONYTALK	WAVE_FORMAT_MALDEN_PHONYTALK
Gsm	WAVE_FORMAT_GSM
G729	WAVE_FORMAT_G729
G723	WAVE_FORMAT_G723
Acelp	WAVE_FORMAT_ACELP
RawAac	WAVE_FORMAT_RAW_AAC1
WAVE_FORMAT_RHETOREX_ADPCM	WAVE_FORMAT_RHETOREX_ADPCM
WAVE_FORMAT_IRAT	WAVE_FORMAT_IRAT
WAVE_FORMAT_VIVO_G723	WAVE_FORMAT_VIVO_G723
WAVE_FORMAT_VIVO_SIREN	WAVE_FORMAT_VIVO_SIREN
WAVE_FORMAT_DIGITAL_G723	WAVE_FORMAT_DIGITAL_G723
WAVE_FORMAT_SANYO_LD_ADPCM	WAVE_FORMAT_SANYO_LD_ADPCM
WAVE_FORMAT_SIPROLAB_ACEPLNET	WAVE_FORMAT_SIPROLAB_ACEPLNET
WAVE_FORMAT_SIPROLAB_ACELP4800	WAVE_FORMAT_SIPROLAB_ACELP4800
WAVE_FORMAT_SIPROLAB_ACELP8V3	WAVE_FORMAT_SIPROLAB_ACELP8V3
WAVE_FORMAT_SIPROLAB_G729	WAVE_FORMAT_SIPROLAB_G729
WAVE_FORMAT_SIPROLAB_G729A	WAVE_FORMAT_SIPROLAB_G729A
WAVE_FORMAT_SIPROLAB_KELVIN	WAVE_FORMAT_SIPROLAB_KELVIN
WAVE_FORMAT_G726ADPCM	WAVE_FORMAT_G726ADPCM
WAVE_FORMAT_QUALCOMM_PUREVOICE	WAVE_FORMAT_QUALCOMM_PUREVOICE
WAVE_FORMAT_QUALCOMM_HALFRATE	WAVE_FORMAT_QUALCOMM_HALFRATE
WAVE_FORMAT_TUBGSM	WAVE_FORMAT_TUBGSM
WAVE_FORMAT_MSAUDIO1	WAVE_FORMAT_MSAUDIO1
WindowsMediaAudio	Windows Media Audio, WAVE_FORMAT_WMAUDIO2, Microsoft Corporation
WindowsMediaAudioProfessional	Windows Media Audio Professional WAVE_FORMAT_WMAUDIO3, Microsoft Corporation
WindowsMediaAudioLossless	Windows Media Audio Lossless, WAVE_FORMAT_WMAUDIO_LOSSLESS
WindowsMediaAudioSpdif	Windows Media Audio Professional over SPDIF WAVE_FORMAT_WMASPDIF (0x0164)
WAVE_FORMAT_UNISYS_NAP_ADPCM	WAVE_FORMAT_UNISYS_NAP_ADPCM
WAVE_FORMAT_UNISYS_NAP_ULAW	WAVE_FORMAT_UNISYS_NAP_ULAW
WAVE_FORMAT_UNISYS_NAP_ALAW	WAVE_FORMAT_UNISYS_NAP_ALAW
WAVE_FORMAT_UNISYS_NAP_16K	WAVE_FORMAT_UNISYS_NAP_16K
WAVE_FORMAT_CREATIVE_ADPCM	WAVE_FORMAT_CREATIVE_ADPCM
WAVE_FORMAT_CREATIVE_FASTSPEECH8	WAVE_FORMAT_CREATIVE_FASTSPEECH8
WAVE_FORMAT_CREATIVE_FASTSPEECH10	WAVE_FORMAT_CREATIVE_FASTSPEECH10
WAVE_FORMAT_UHER_ADPCM	WAVE_FORMAT_UHER_ADPCM

Enumerator

WAVE_FORMAT_QUARTERDECK	WAVE_FORMAT_QUARTERDECK
WAVE_FORMAT_ILINK_VC	WAVE_FORMAT_ILINK_VC
WAVE_FORMAT_RAW_SPORT	WAVE_FORMAT_RAW_SPORT
WAVE_FORMAT_ESST_AC3	WAVE_FORMAT_ESST_AC3
WAVE_FORMAT_IPI_HSX	WAVE_FORMAT_IPI_HSX
WAVE_FORMAT_IPI_RPELP	WAVE_FORMAT_IPI_RPELP
WAVE_FORMAT_CS2	WAVE_FORMAT_CS2
WAVE_FORMAT_SONY_SCX	WAVE_FORMAT_SONY_SCX
WAVE_FORMAT_FM_TOWNS_SND	WAVE_FORMAT_FM_TOWNS_SND
WAVE_FORMAT_BTV_DIGITAL	WAVE_FORMAT_BTV_DIGITAL
WAVE_FORMAT_QDESIGN_MUSIC	WAVE_FORMAT_QDESIGN_MUSIC
WAVE_FORMAT_VME_VMPCM	WAVE_FORMAT_VME_VMPCM
WAVE_FORMAT_TPC	WAVE_FORMAT_TPC
WAVE_FORMAT_OLIGSM	WAVE_FORMAT_OLIGSM
WAVE_FORMAT_OLIADPCM	WAVE_FORMAT_OLIADPCM
WAVE_FORMAT_OLICELP	WAVE_FORMAT_OLICELP
WAVE_FORMAT_OLISBC	WAVE_FORMAT_OLISBC
WAVE_FORMAT_OLIOPR	WAVE_FORMAT_OLIOPR
WAVE_FORMAT_LH_CODEC	WAVE_FORMAT_LH_CODEC
WAVE_FORMAT_NORRIS	WAVE_FORMAT_NORRIS
WAVE_FORMAT_SOUNDSPACE_MUSICOMPR← ESS	WAVE_FORMAT_SOUNDSPACE_MUSICOMPRES← SS
MPEG_ADTS_AAC	Advanced Audio Coding (AAC) audio in Audio Data Transport Stream (ADTS) format. The format block is a WAVEFORMATEX structure with wFormatTag equal to WAVE_FORMAT_MPEG_ADTS_AAC. The WAVEFORMATEX structure specifies the core AAC-LC sample rate and number of channels, prior to applying spectral band replication (SBR) or parametric stereo (PS) tools, if present. No additional data is required after the WAVEFORMATEX structure. http://msdn.microsoft.com/en-us/library/dd317599%28VS.85%29.aspx
MPEG_RAW_AAC	MPEG_RAW_AAC Source wmCodec.h
MPEG_LOAS	MPEG-4 audio transport stream with a synchronization layer (LOAS) and a multiplex layer (LATM). The format block is a WAVEFORMATEX structure with wFormatTag equal to WAVE_FORMAT_MPEG_LOAS. See . The WAVEFORMATEX structure specifies the core AAC-LC sample rate and number of channels, prior to applying spectral SBR or PS tools, if present. No additional data is required after the WAVEFORMATEX structure.
NOKIA_MPEG_ADTS_AAC	NOKIA_MPEG_ADTS_AAC Source wmCodec.h
NOKIA_MPEG_RAW_AAC	NOKIA_MPEG_RAW_AAC Source wmCodec.h
VODAFONE_MPEG_ADTS_AAC	VODAFONE_MPEG_ADTS_AAC Source wmCodec.h
VODAFONE_MPEG_RAW_AAC	VODAFONE_MPEG_RAW_AAC Source wmCodec.h

Enumerator

MPEG_HEAAC	High-Efficiency Advanced Audio Coding (HE-AAC) stream. The format block is an HEAACWAVEFORMAT structure. See .
WAVE_FORMAT_DVM	WAVE_FORMAT_DVM
Vorbis1	WAVE_FORMAT_VORBIS1 "Og" Original stream compatible
Vorbis2	WAVE_FORMAT_VORBIS2 "Pg" Have independent header
Vorbis3	WAVE_FORMAT_VORBIS3 "Qg" Have no codebook header
Vorbis1P	WAVE_FORMAT_VORBIS1P "og" Original stream compatible
Vorbis2P	WAVE_FORMAT_VORBIS2P "pg" Have independent headere
Vorbis3P	WAVE_FORMAT_VORBIS3P "qg" Have no codebook header
WAVE_FORMAT_RAW_AAC1	Raw AAC1
WAVE_FORMAT_WMAVOICE9	Windows Media Audio Voice (WMA Voice)
Extensible	Extensible
WAVE_FORMAT_DEVELOPMENT	WAVE_FORMAT_DEVELOPMENT
WAVE_FORMAT_FLAC	FLAC

3.1.1.2 ChannelMask

enum `ChannelMask` [strong]

Channelmask used by [WaveFormatExtensible](#). For more information see [http://msdn.microsoft.com/en-us/library/windows/desktop/dd757714\(v=vs.85\).aspx](http://msdn.microsoft.com/en-us/library/windows/desktop/dd757714(v=vs.85).aspx)

Enumerator

SpeakerFrontLeft	Front left speaker.
SpeakerFrontRight	Front right speaker.
SpeakerFrontCenter	Front center speaker.
SpeakerLowFrequency	Low frequency speaker.
SpeakerBackLeft	Back left speaker.
SpeakerBackRight	Back right speaker.
SpeakerFrontLeftOfCenter	Front left of center speaker.
SpeakerFrontRightOfCenter	Front right of center speaker.
SpeakerBackCenter	Back center speaker.
SpeakerSideLeft	Side left speaker.
SpeakerSideRight	Side right speaker.
SpeakerTopCenter	Top center speaker.
SpeakerTopFrontLeft	Top front left speaker.
SpeakerTopFrontCenter	Top front center speaker.
SpeakerTopFrontRight	Top front right speaker.
SpeakerTopBackLeft	Top back left speaker.
SpeakerTopBackCenter	Top back center speaker.
SpeakerTopBackRight	Top back right speaker.

3.2 CSCore.Codecs Namespace Reference

3.3 CSCore.Codecs.WAV Namespace Reference

Classes

- class [WaveWriter](#)
Encoder for wave files.

3.4 Photon Namespace Reference

3.5 Photon.Voice Namespace Reference

Classes

- class [AudioDesc](#)
- class [AudioInChangeNotifier](#)
- class [AudioInEnumerator](#)
Enumerates microphones available on device.
- class [AudioSyncBuffer](#)
- class [AudioUtil](#)
Collection of Audio Utility functions and classes.
- class [BufferReaderPushAdapter](#)
Simple [BufferReaderPushAdapterBase](#) implementation using a single buffer, using synchronous [LocalVoice.PushData](#)
- class [BufferReaderPushAdapterAsyncPool](#)
[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#).
- class [BufferReaderPushAdapterAsyncPoolCopy](#)
[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#) and data copy.
- class [BufferReaderPushAdapterAsyncPoolFloatToShort](#)
[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#), converting float samples to short.
- class [BufferReaderPushAdapterAsyncPoolShortToFloat](#)
[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#), converting short samples to float.
- class [BufferReaderPushAdapterBase](#)
Adapter base class to move data by reading from [IDataReader.Read](#) and pushing to [LocalVoice](#).
- class [FactoryPrimitiveArrayPool](#)
[PrimitiveArrayPool<T>](#) as wrapped in object factory interface.
- class [FactoryReusableArray](#)
Array factory returning the same array instance as long as it requested with the same array length. If length changes, new array instance created.
- class [FrameOut](#)
- class [Framer](#)
Utility class to re-frame audio packets.
- interface [IAudioDesc](#)
Audio Source interface.
- interface [IAudioOut](#)
- interface [IAudioPusher](#)

- Audio Pusher interface.*
 - interface [IAudioReader](#)
 - Audio Reader interface.*
 - interface [IDataReader](#)
 - Interface for pulling data, in case this is more appropriate than pushing it.*
 - interface [IDecoder](#)
 - Generic decoder interface.*
 - interface [IDecoderDirect](#)
 - Interface for an decoder which outputs data via explicit call.*
 - interface [IDecoderQueuedOutputImageNative](#)
 - interface [IEncoder](#)
 - Generic encoder interface.*
 - interface [IEncoderDirect](#)
 - Interface for an encoder which consumes input data via explicit call.*
 - interface [ILocalVoiceAudio](#)
 - Interface for an outgoing audio stream.*
 - interface [ILogger](#)
 - class [ImageBufferInfo](#)
 - class [ImageBufferNative](#)
 - class [ImageBufferNativeAlloc](#)
 - class [ImageBufferNativeGCHandleSinglePlane](#)
 - class [ImageBufferNativePool](#)
 - struct [ImageInputBuf](#)
 - struct [ImageOutputBuf](#)
 - interface [IProcessor](#)
 - Audio Processor interface.*
 - interface [IResettable](#)
 - interface [IServiceable](#)
 - Interface for classes that want their [Service\(\)](#) function to be called regularly in the context of a [LocalVoice](#).*
 - interface [IVoiceTransport](#)
 - class [LoadBalancingFrontend](#)
 - class [LoadBalancingTransport](#)
 - Extends [LoadBalancingClient](#) with audio streaming functionality.*
 - class [LoadBalancingTransport2](#)
 - Variant of [LoadBalancingTransport](#). Aims to be non-alloc at the cost of breaking compatibility with older clients.*
 - class [LocalVoice](#)
 - Represents outgoing data stream.*
 - class [LocalVoiceAudio](#)
 - Outgoing audio stream.*
 - class [LocalVoiceAudioDummy](#)
 - Dummy [LocalVoiceAudio](#)*
 - class [LocalVoiceAudioFloat](#)
 - Specialization of [LocalVoiceAudio](#) for float audio*
 - class [LocalVoiceAudioShort](#)
 - Specialization of [LocalVoiceAudio](#) for short audio*
 - class [LocalVoiceFramed](#)
 - Typed re-framing [LocalVoice](#)*
 - class [LocalVoiceFramedBase](#)
 - Typed re-framing [LocalVoice](#)*
 - class [MonoPInvokeCallbackAttribute](#)
 - interface [ObjectFactory](#)
 - Uniform interface to [ObjectPool< TType, TInfo>](#) and single reusable object.*

- class [ObjectPool](#)
Generic Pool to re-use objects of a certain type (TType) that optionally match a certain property or set of properties (TInfo).
- class [OpusCodec](#)
- class **PhotonTransportProtocol**
- class [Platform](#)
- class [PrimitiveArrayPool](#)
Pool of Arrays with components of type T, with [ObjectPool](#) info being the array's size.
- class [RawCodec](#)
- class **RemoteVoice**
- class [RemoteVoiceInfo](#)
Information about a remote voice (incoming stream).
- struct [RemoteVoiceOptions](#)
Event Actions and other options for a remote voice (incoming stream).
- class **SpacingProfile**
- class [UnsupportedCodecException](#)
Exception thrown if an unsupported codec is encountered.
- class [UnsupportedSampleTypeException](#)
Exception thrown if an unsupported audio sample type is encountered.
- class [VoiceClient](#)
[Voice](#) client interact with other clients on network via [IVoiceTransport](#).
- class [VoiceEvent](#)
- struct [VoiceInfo](#)
Describes stream properties.
- class [WebRTCAudioLib](#)
- class [WebRTCAudioProcessor](#)

Enumerations

- enum [AudioSampleType](#)
The type of samples used for audio processing.
- enum **FrameFlags** : byte
- enum [Codec](#)
Enum for Media Codecs supported by PhotonVoice.
- enum **ImageFormat**
- enum **Rotation**
- enum **Flip**

3.5.1 Enumeration Type Documentation

3.5.1.1 AudioSampleType

```
enum AudioSampleType [strong]
```

The type of samples used for audio processing.

3.5.1.2 Codec

```
enum Codec [strong]
```

Enum for Media Codecs supported by PhotonVoice.

Transmitted in [VoiceInfo](#). Do not change the values of this Enum!

Enumerator

AudioOpus	OPUS audio
-----------	------------

3.6 Photon.Voice.IOS Namespace Reference

Classes

- struct [AudioSessionParameters](#)
- class [AudioSessionParametersPresets](#)

Enumerations

- enum [AudioSessionCategory](#)
- enum [AudioSessionMode](#)
- enum [AudioSessionCategoryOption](#)

3.6.1 Enumeration Type Documentation

3.6.1.1 AudioSessionCategory

enum [AudioSessionCategory](#) [strong]

Enumerator

Ambient	Use this category for background sounds such as rain, car engine noise, etc. Mixes with other music. API_AVAILABLE(ios(3.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
SoloAmbient	Use this category for background sounds. Other music will stop playing. API_AVAILABLE(ios(3.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
Playback	Use this category for music tracks. API_AVAILABLE(ios(3.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
Record	Use this category when recording audio. API_AVAILABLE(ios(3.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
PlayAndRecord	Use this category when recording and playing back audio. API_AVAILABLE(ios(3.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
AudioProcessing	Use this category when using a hardware codec or signal processor while not playing or recording audio. API_DEPRECATED("No longer supported", ios(3.0, 10.0)) API_UNAVAILABLE(watchos, tvos) API_UNAVAILABLE(macos);

Enumerator

MultiRoute	<p>Use this category to customize the usage of available audio accessories and built-in audio hardware. For example, this category provides an application with the ability to use an available USB output and headphone output simultaneously for separate, distinct streams of audio data. Use of this category by an application requires a more detailed knowledge of, and interaction with, the capabilities of the available audio routes. May be used for input, output, or both. Note that not all output types and output combinations are eligible for multi-route. Input is limited to the last-in input port. Eligible inputs consist of the following: AVAudioSessionPortUSBAudio, AVAudioSessionPortHeadsetMic, and AVAudioSessionPortBuiltInMic.</p> <p>Eligible outputs consist of the following: AVAudioSessionPortUSBAudio, AVAudioSessionPortLineOut, AVAudioSessionPortHeadphones, AVAudioSessionPortHDMI, and AVAudioSessionPortBuiltInSpeaker.</p> <p>Note that AVAudioSessionPortBuiltInSpeaker is only allowed to be used when there are no other outputs connected. API_AVAILABLE(ios(6.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);</p>
------------	---

3.6.1.2 AudioSessionCategoryOption

```
enum AudioSessionCategoryOption [strong]
```

Enumerator

MixWithOthers	<p>This allows an application to set whether or not other active audio apps will be interrupted or mixed with when your app's audio session goes active. The typical cases are: (1) AVAudioSessionCategoryPlayAndRecord or AVAudioSessionCategoryMultiRoute this will default to false, but can be set to true. This would allow other applications to play in the background while an app had both audio input and output enabled (2) AVAudioSessionCategoryPlayback this will default to false, but can be set to true. This would allow other applications to play in the background, but an app will still be able to play regardless of the setting of the ringer switch (3) Other categories this defaults to false and cannot be changed (that is, the mix with others setting of these categories cannot be overridden. An application must be prepared for setting this property to fail as behaviour may change in future releases. If an application changes their category, they should reassert the option (it is not sticky across category changes). MixWithOthers is only valid with AVAudioSessionCategoryPlayAndRecord, AVAudioSessionCategoryPlayback, and AVAudioSessionCategoryMultiRoute</p>
DuckOthers	<p>This allows an application to set whether or not other active audio apps will be ducked when when your app's audio session goes active. An example of this is the Nike app, which provides periodic updates to its user (it reduces the volume of any music currently being played while it provides its status). This defaults to off. Note that the other audio will be ducked for as long as the current session is active. You will need to deactivate your audio session when you want full volume playback of the other audio. If your category is AVAudioSessionCategoryPlayback, AVAudioSessionCategoryPlayAndRecord, or AVAudioSessionCategoryMultiRoute, by default the audio session will be non-mixable and non-ducking. Setting this option will also make your category mixable with others (AVAudioSessionCategoryOptionMixWithOthers will be set). DuckOthers is only valid with AVAudioSessionCategoryAmbient, AVAudioSessionCategoryPlayAndRecord, AVAudioSessionCategoryPlayback, and AVAudioSessionCategoryMultiRoute</p>

Enumerator

AllowBluetooth	<p>This allows an application to change the default behaviour of some audio session categories with regards to showing bluetooth Hands-Free Profile (HFP) devices as available routes. The current category behavior is: (1) AVAudioSessionCategoryPlayAndRecord this will default to false, but can be set to true. This will allow a paired bluetooth HFP device to show up as an available route for input, while playing through the category-appropriate output (2) AVAudioSessionCategoryRecord this will default to false, but can be set to true. This will allow a paired bluetooth HFP device to show up as an available route for input (3) Other categories this defaults to false and cannot be changed (that is, enabling bluetooth for input in these categories is not allowed) An application must be prepared for setting this option to fail as behaviour may change in future releases. If an application changes their category or mode, they should reassert the override (it is not sticky across category and mode changes). AllowBluetooth is only valid with AVAudioSessionCategoryRecord and AVAudioSessionCategoryPlayAndRecord</p>
DefaultToSpeaker	<p>This allows an application to change the default behaviour of some audio session categories with regards to the audio route. The current category behavior is: (1) AVAudioSessionCategoryPlayAndRecord category this will default to false, but can be set to true. this will route to Speaker (instead of Receiver) when no other audio route is connected. (2) Other categories this defaults to false and cannot be changed (that is, the default to speaker setting of these categories cannot be overridden An application must be prepared for setting this property to fail as behaviour may change in future releases. If an application changes their category, they should reassert the override (it is not sticky across category and mode changes). DefaultToSpeaker is only valid with AVAudioSessionCategoryPlayAndRecord</p>

3.6.1.3 AudioSessionMode

```
enum AudioSessionMode [strong]
```

Enumerator

Default	Modes modify the audio category in order to introduce behavior that is tailored to the specific use of audio within an application. Available in iOS 5.0 and greater. The default mode API_AVAILABLE(ios(5.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
VoiceChat	Only valid with AVAudioSessionCategoryPlayAndRecord. Appropriate for Voice over IP (VoIP) applications. Reduces the number of allowable audio routes to be only those that are appropriate for VoIP applications and may engage appropriate system-supplied signal processing. Has the side effect of setting AVAudioSessionCategoryOptionAllowBluetooth API_AVAILABLE(ios(5.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
VideoRecording	Only valid with AVAudioSessionCategoryPlayAndRecord or AVAudioSessionCategoryRecord. Modifies the audio routing options and may engage appropriate system-supplied signal processing. API_AVAILABLE(ios(5.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
Measurement	Appropriate for applications that wish to minimize the effect of system-supplied signal processing for input and/or output audio signals. API_AVAILABLE(ios(5.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);
MoviePlayback	Engages appropriate output signal processing for movie playback scenarios. Currently only applied during playback over built-in speaker. API_AVAILABLE(ios(6.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);

Enumerator

VideoChat	Only valid with <code>kAudioSessionCategory_PlayAndRecord</code> . Reduces the number of allowable audio routes to be only those that are appropriate for video chat applications. May engage appropriate system-supplied signal processing. Has the side effect of setting <code>AVAudioSessionCategoryOptionAllowBluetooth</code> and <code>AVAudioSessionCategoryOptionDefaultToSpeaker</code> . <code>API_AVAILABLE(ios(7.0), watchos(2.0), tvos(9.0)) API_UNAVAILABLE(macos);</code>
-----------	---

3.7 Photon.Voice.PUN Namespace Reference

Classes

- class [PhotonVoiceNetwork](#)

This class can be used to automatically sync client states between [PUN](#) and [Voice](#). It also sets a custom [PUN Speaker](#) factory to find the [Speaker](#) component for a character's voice. For this to work attach a [PhotonVoiceView](#) next to the [PhotonView](#) of your player's prefab.

- class [PhotonVoiceView](#)

Component that should be attached to a networked [PUN](#) prefab that has [PhotonView](#). It will bind remote [Recorder](#) with local [Speaker](#) of the same networked prefab. This component makes automatic voice stream routing easy for players' characters/avatars.

3.8 Photon.Voice.Unity Namespace Reference

Classes

- class [AudioClipWrapper](#)
- class [AudioOutCapture](#)
- interface [ILoggable](#)
- interface [ILoggableDependent](#)
- class [IOSAudioForceToSpeaker](#)
- class [Logger](#)
- class [MicWrapper](#)
- class [MicWrapperPusher](#)
- struct [NativeAndroidMicrophoneSettings](#)
- class [PhotonVoiceCreatedParams](#)
- struct [PlaybackDelaySettings](#)

Playback delay configuration container.

- class [Recorder](#)

Component representing outgoing audio stream in scene.

- class [RemoteVoiceLink](#)
- class [Speaker](#)

Component representing remote audio stream in local scene.

- class [UnityAndroidAudioInAEC](#)
- class [UnityAudioOut](#)
- class [UnityMicrophone](#)

A wrapper around `UnityEngine.Microphone` to be able to safely use `Microphone` and compile for WebGL.

- class [VoiceComponent](#)
- class [VoiceConnection](#)

Component that represents a client voice connection to [Photon](#) Servers.

- class [VoiceLogger](#)
- class [WebRtcAudioDsp](#)

3.9 Photon.Voice.Unity.UtilityScripts Namespace Reference

Classes

- class [ConnectAndJoin](#)
- class [MicAmplifier](#)
- class [MicAmplifierFloat](#)
- class [MicAmplifierShort](#)
- class [PhotonVoiceLagSimulationGui](#)
- class [PhotonVoiceStatsGui](#)

Basic GUI to show traffic and health statistics of the connection to [Photon](#), toggled by shift+tab.

- class [SaveIncomingStreamToFile](#)
- class [SaveOutgoingStreamToFile](#)
- class [TestTone](#)
- class [ToneAudioReader](#)

3.10 POpusCodec Namespace Reference

Classes

- class [OpusDecoder](#)
- class [OpusEncoder](#)
- class [OpusException](#)
- class [OpusLib](#)
- class [Wrapper](#)

3.11 POpusCodec.Enums Namespace Reference

Enumerations

- enum [Bandwidth](#) : int
- enum [Channels](#) : int
- enum **Complexity** : int
- enum [Delay](#)

Using a duration of less than 10 ms will prevent the encoder from using the LPC or hybrid modes.

- enum **ForceChannels** : int
- enum [OpusApplicationType](#) : int
- enum **OpusStatusCode** : int
- enum **SamplingRate** : int
- enum [SignalHint](#) : int

3.11.1 Enumeration Type Documentation

3.11.1.1 Bandwidth

```
enum Bandwidth : int [strong]
```

Enumerator

Narrowband	Up to 4Khz
Mediumband	Up to 6Khz
Wideband	Up to 8Khz
SuperWideband	Up to 12Khz
Fullband	Up to 20Khz (High Definition)

3.11.1.2 Channels

```
enum Channels : int [strong]
```

Enumerator

Mono	1 Channel
Stereo	2 Channels

3.11.1.3 Delay

```
enum Delay [strong]
```

Using a duration of less than 10 ms will prevent the encoder from using the LPC or hybrid modes.

Enumerator

Delay2dot5ms	2.5ms
Delay5ms	5ms
Delay10ms	10ms
Delay20ms	20ms
Delay40ms	40ms
Delay60ms	60ms

3.11.1.4 OpusApplicationType

```
enum OpusApplicationType : int [strong]
```

Enumerator

Voip	Gives best quality at a given bitrate for voice signals. It enhances the input signal by high-pass filtering and emphasizing formants and harmonics. Optionally it includes in-band forward error correction to protect against packet loss. Use this mode for typical VoIP applications. Because of the enhancement, even at high bitrates the output may sound different from the input.
Audio	Gives best quality at a given bitrate for most non-voice signals like music. Use this mode for music and mixed (music/voice) content, broadcast, and applications requiring less than 15 ms of coding delay.
RestrictedLowDelay	Configures low-delay mode that disables the speech-optimized mode in exchange for slightly reduced delay.

3.11.1.5 SignalHint

```
enum SignalHint : int [strong]
```

Enumerator

Auto	(default)
Voice	Bias thresholds towards choosing LPC or Hybrid modes
Music	Bias thresholds towards choosing MDCT modes.

Chapter 4

Class Documentation

4.1 AudioClipWrapper Class Reference

Inherits [IAudioReader< float >](#).

Public Member Functions

- **AudioClipWrapper** (AudioClip audioClip)
- bool **Read** (float[] buffer)
- void **Dispose** ()

Properties

- bool **Loop** [get, set]
- int **SamplingRate** [get]
- int **Channels** [get]
- string **Error** [get]

4.2 AudioDesc Class Reference

Inherits [IAudioDesc](#).

Public Member Functions

- **AudioDesc** (int samplingRate, int channels, string error)
- void **Dispose** ()

Properties

- int **SamplingRate** [get]
- int **Channels** [get]
- string **Error** [get]

4.3 **AudiInChangeNotifier** Class Reference

Inherits `IDisposable`.

Public Member Functions

- **AudiInChangeNotifier** (Action callback, [ILogger](#) logger)
- void **Dispose** ()

Public Attributes

- readonly bool **IsSupported** = false

Properties

- string **Error** [get]

4.4 **AudiInEnumerator** Class Reference

Enumerates microphones available on device.

Inherits `IDisposable`.

Public Member Functions

- **AudiInEnumerator** ([ILogger](#) logger)
- void **Refresh** ()
- string **NameAtIndex** (int i)
- int **IDAtIndex** (int i)
- bool **IDIsValid** (int id)
- void **Dispose** ()

Public Attributes

- readonly bool **IsSupported** = false

Properties

- string **Error** [get]
- int **Count** [get]

4.4.1 Detailed Description

Enumerates microphones available on device.

4.5 AudioOutCapture Class Reference

Inherits MonoBehaviour.

Events

- Action< float[], int > **OnAudioFrame**

4.6 AudioSessionParameters Struct Reference

Public Member Functions

- int **CategoryOptionsToInt** ()
- override string **ToString** ()

Public Attributes

- [AudioSessionCategory](#) **Category**
- [AudioSessionMode](#) **Mode**
- [AudioSessionCategoryOption](#)[] **CategoryOptions**

4.7 AudioSessionParametersPresets Class Reference

Static Public Attributes

- static [AudioSessionParameters](#) **Game**
- static [AudioSessionParameters](#) **VoIP**

4.7.1 Member Data Documentation

4.7.1.1 Game

```
AudioSessionParameters Game [static]
```

Initial value:

```
= new AudioSessionParameters()
{
    Category = AudioSessionCategory.PlayAndRecord,
    Mode = AudioSessionMode.Default,
    CategoryOptions = new AudioSessionCategoryOption[] {
        AudioSessionCategoryOption.DefaultToSpeaker, AudioSessionCategoryOption.AllowBluetooth }
}
```

4.7.1.2 VoIP

`AudioSessionParameters VoIP [static]`

Initial value:

```
= new AudioSessionParameters()
{
    Category = AudioSessionCategory.PlayAndRecord,
    Mode = AudioSessionMode.VoiceChat,

    CategoryOptions = new AudioSessionCategoryOption[] { AudioSessionCategoryOption.AllowBluetooth }
}
```

4.8 AudioSubTypes Class Reference

Defines [AudioSubTypes](#) and provides methods to convert between [AudioEncoding](#)-values and [AudioSubTypes](#)-values.

Static Public Member Functions

- static [AudioEncoding](#) [EncodingFromSubType](#) (Guid audioSubType)
Converts a [AudioSubTypes](#)-value to a [AudioEncoding](#)-value.
- static Guid [SubTypeFromEncoding](#) ([AudioEncoding](#) audioEncoding)
Converts a [AudioEncoding](#) value to a [AudioSubTypes](#)-value.

Static Public Attributes

- static readonly Guid [MediaTypeAudio](#) = new Guid("73647561-0000-0010-8000-00AA00389B71")
The Major Type for Audio media types.
- static readonly Guid [Unknown](#) = new Guid((short)AudioEncoding.Unknown & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_UNKNOWN, Microsoft Corporation
- static readonly Guid [Pcm](#) = new Guid((short)AudioEncoding.Pcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_PCM Microsoft Corporation
- static readonly Guid [Adpcm](#) = new Guid((short)AudioEncoding.Adpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_ADPCM Microsoft Corporation
- static readonly Guid [IeeeFloat](#) = new Guid((short)AudioEncoding.IeeeFloat & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_IEEE_FLOAT Microsoft Corporation
- static readonly Guid [Vseip](#) = new Guid((short)AudioEncoding.Vseip & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_VSELP Compaq Computer Corp.
- static readonly Guid [IbmCvvd](#) = new Guid((short)AudioEncoding.IbmCvvd & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_IBM_CVSD IBM Corporation
- static readonly Guid [ALaw](#) = new Guid((short)AudioEncoding.ALaw & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_ALAW Microsoft Corporation

- static readonly Guid [MuLaw](#) = new Guid((short)AudioEncoding.MuLaw & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_MULAW Microsoft Corporation
- static readonly Guid [Dts](#) = new Guid((short)AudioEncoding.Dts & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DTS Microsoft Corporation
- static readonly Guid [Drm](#) = new Guid((short)AudioEncoding.Drm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DRM Microsoft Corporation
- static readonly Guid [WmaVoice9](#) = new Guid((short)AudioEncoding.WmaVoice9 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_WMAVOICE9
- static readonly Guid [OkiAdpcm](#) = new Guid((short)AudioEncoding.OkiAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_OKI_ADPCM OKI
- static readonly Guid [DviAdpcm](#) = new Guid((short)AudioEncoding.DviAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DVI_ADPCM Intel Corporation
- static readonly Guid [ImaAdpcm](#) = new Guid((short)AudioEncoding.ImaAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_IMA_ADPCM Intel Corporation
- static readonly Guid [MediaspaceAdpcm](#) = new Guid((short)AudioEncoding.MediaspaceAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_MEDIASPACE_ADPCM Videologic
- static readonly Guid [SierraAdpcm](#) = new Guid((short)AudioEncoding.SierraAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SIERRA_ADPCM Sierra Semiconductor Corp
- static readonly Guid [G723Adpcm](#) = new Guid((short)AudioEncoding.G723Adpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_G723_ADPCM Antex Electronics Corporation
- static readonly Guid [DigiStd](#) = new Guid((short)AudioEncoding.DigiStd & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DIGISTD DSP Solutions, Inc.
- static readonly Guid [DigiFix](#) = new Guid((short)AudioEncoding.DigiFix & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DIGIFIX DSP Solutions, Inc.
- static readonly Guid [DialogicOkiAdpcm](#) = new Guid((short)AudioEncoding.DialogicOkiAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DIALOGIC_OKI_ADPCM Dialogic Corporation
- static readonly Guid [MediaVisionAdpcm](#) = new Guid((short)AudioEncoding.MediaVisionAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_MEDIAVISION_ADPCM Media Vision, Inc.
- static readonly Guid [CUCodec](#) = new Guid((short)AudioEncoding.CUCodec & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_CU_CODEC Hewlett-Packard Company
- static readonly Guid [YamahaAdpcm](#) = new Guid((short)AudioEncoding.YamahaAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_YAMAHA_ADPCM Yamaha Corporation of America
- static readonly Guid [SonarC](#) = new Guid((short)AudioEncoding.SonarC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SONARC Speech Compression
- static readonly Guid [DspGroupTrueSpeech](#) = new Guid((short)AudioEncoding.DspGroupTrueSpeech & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

- WAVE_FORMAT_DSPGROUP_TRUESPEECH DSP Group, Inc*
- static readonly Guid [EchoSpeechCorporation1](#) = new Guid((short)AudioEncoding.EchoSpeechCorporation1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_ECHOSC1 Echo Speech Corporation*
- static readonly Guid [AudioFileAf36](#) = new Guid((short)AudioEncoding.AudioFileAf36 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_AUDIOFILE_AF36, Virtual Music, Inc.*
- static readonly Guid [Aptx](#) = new Guid((short)AudioEncoding.Aptx & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_APTX Audio Processing Technology*
- static readonly Guid [AudioFileAf10](#) = new Guid((short)AudioEncoding.AudioFileAf10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_AUDIOFILE_AF10, Virtual Music, Inc.*
- static readonly Guid [Prosody1612](#) = new Guid((short)AudioEncoding.Prosody1612 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_PROSODY_1612, Aculab plc*
- static readonly Guid [Lrc](#) = new Guid((short)AudioEncoding.Lrc & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_LRC, Merging Technologies S.A.*
- static readonly Guid [DolbyAc2](#) = new Guid((short)AudioEncoding.DolbyAc2 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_DOLBY_AC2, Dolby Laboratories*
- static readonly Guid [Gsm610](#) = new Guid((short)AudioEncoding.Gsm610 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_GSM610, Microsoft Corporation*
- static readonly Guid [MsnAudio](#) = new Guid((short)AudioEncoding.MsnAudio & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_MSNAUDIO, Microsoft Corporation*
- static readonly Guid [AntexAdpcme](#) = new Guid((short)AudioEncoding.AntexAdpcme & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_ANTEX_ADPCME, Antex Electronics Corporation*
- static readonly Guid [ControlResVqlpc](#) = new Guid((short)AudioEncoding.ControlResVqlpc & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_CONTROL_RES_VQLPC, Control Resources Limited*
- static readonly Guid [DigiReal](#) = new Guid((short)AudioEncoding.DigiReal & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_DIGIREAL, DSP Solutions, Inc.*
- static readonly Guid [DigiAdpcm](#) = new Guid((short)AudioEncoding.DigiAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_DIGIADPCM, DSP Solutions, Inc.*
- static readonly Guid [ControlResCr10](#) = new Guid((short)AudioEncoding.ControlResCr10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_CONTROL_RES_CR10, Control Resources Limited*
- static readonly Guid [WAVE_FORMAT_NMS_VBXADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_NMS_VBXADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_NMS_VBXADPCM*
- static readonly Guid [WAVE_FORMAT_CS_IMAADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_CS_IMAADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_CS_IMAADPCM*
- static readonly Guid [WAVE_FORMAT_ECHOSC3](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ECHOSC3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_ECHOSC3*

- static readonly Guid [WAVE_FORMAT_ROCKWELL_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ROCKWELL_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_ROCKWELL_ADPCM
- static readonly Guid [WAVE_FORMAT_ROCKWELL_DIGITALK](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ROCKWELL_DIGITALK & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_ROCKWELL_DIGITALK
- static readonly Guid [WAVE_FORMAT_XEBEC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_XEBEC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_XEBEC
- static readonly Guid [WAVE_FORMAT_G721_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_G721_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_G721_ADPCM
- static readonly Guid [WAVE_FORMAT_G728_CELP](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_G728_CELP & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_G728_CELP
- static readonly Guid [WAVE_FORMAT_MSG723](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_MSG723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_MSG723
- static readonly Guid [Mpeg](#) = new Guid((short)AudioEncoding.Mpeg & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_MPEG, Microsoft Corporation
- static readonly Guid [WAVE_FORMAT_RT24](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_RT24 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_RT24
- static readonly Guid [WAVE_FORMAT_PAC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_PAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_PAC
- static readonly Guid [MpegLayer3](#) = new Guid((short)AudioEncoding.MpegLayer3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_MPEGLAYER3, ISO/MPEG Layer3 Format Tag
- static readonly Guid [WAVE_FORMAT_LUCENT_G723](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_LUCENT_G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_LUCENT_G723
- static readonly Guid [WAVE_FORMAT_CIRRUS](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_CIRRUS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_CIRRUS
- static readonly Guid [WAVE_FORMAT_ESPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ESPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_ESPCM
- static readonly Guid [WAVE_FORMAT_VOXWARE](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_VOXWARE
- static readonly Guid [WAVE_FORMAT_CANOPUS_ATRAC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_CANOPUS_ATRAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_CANOPUS_ATRAC
- static readonly Guid [WAVE_FORMAT_G726_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_G726_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_G726_ADPCM
- static readonly Guid [WAVE_FORMAT_G722_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_G722_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

- WAVE_FORMAT_G722_ADPCM*

 - static readonly Guid [WAVE_FORMAT_DSAT_DISPLAY](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_DSAT_DISPLAY & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_DSAT_DISPLAY*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_BYTE_ALIGNED](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_BYTE_ALIGNED & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_BYTE_ALIGNED*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_AC8](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC8 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_AC8*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_AC10](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_AC10*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_AC16](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC16 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_AC16*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_AC20](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC20 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_AC20*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_RT24](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_RT24 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_RT24*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_RT29](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_RT29 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_RT29*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_RT29HW](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_RT29HW & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_RT29HW*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_VR12](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_VR12 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_VR12*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_VR18](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_VR18 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_VR18*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_TQ40](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_TQ40 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_TQ40*

 - static readonly Guid [WAVE_FORMAT_SOFTSOUND](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SOFTSOUND & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_SOFTSOUND*

 - static readonly Guid [WAVE_FORMAT_VOXWARE_TQ60](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_TQ60 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_VOXWARE_TQ60*

 - static readonly Guid [WAVE_FORMAT_MSRT24](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_MSRT24 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_MSRT24*

 - static readonly Guid [WAVE_FORMAT_G729A](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_G729A & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_G729A*

 - static readonly Guid [WAVE_FORMAT_MVI_MVI2](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_MVI_MVI2 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

- WAVE_FORMAT_MVI_MVI2*

 - static readonly Guid [WAVE_FORMAT_DF_G726](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_DF_G726 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_DF_G726*

 - static readonly Guid [WAVE_FORMAT_DF_GSM610](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_DF_GSM610 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_DF_GSM610*

 - static readonly Guid [WAVE_FORMAT_ISIAUDIO](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ISIAUDIO & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_ISIAUDIO*

 - static readonly Guid [WAVE_FORMAT_ONLIVE](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ONLIVE & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_ONLIVE*

 - static readonly Guid [WAVE_FORMAT_SBC24](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SBC24 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_SBC24*

 - static readonly Guid [WAVE_FORMAT_DOLBY_AC3_SPDIF](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_DOLBY_AC3_SPDIF & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_DOLBY_AC3_SPDIF*

 - static readonly Guid [WAVE_FORMAT_MEDIASONIC_G723](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_MEDIASONIC_G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_MEDIASONIC_G723*

 - static readonly Guid [WAVE_FORMAT_PROSODY_8KBPS](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_PROSODY_8KBPS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_PROSODY_8KBPS*

 - static readonly Guid [WAVE_FORMAT_ZYXEL_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ZYXEL_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_ZYXEL_ADPCM*

 - static readonly Guid [WAVE_FORMAT_PHILIPS_LPCBB](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_PHILIPS_LPCBB & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_PHILIPS_LPCBB*

 - static readonly Guid [WAVE_FORMAT_PACKED](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_PACKED & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_PACKED*

 - static readonly Guid [WAVE_FORMAT_MALDEN_PHONYTALK](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_MALDEN_PHONYTALK & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_MALDEN_PHONYTALK*

 - static readonly Guid [Gsm](#) = new Guid((short)AudioEncoding.Gsm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_GSM*

 - static readonly Guid [G729](#) = new Guid((short)AudioEncoding.G729 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_G729*

 - static readonly Guid [G723](#) = new Guid((short)AudioEncoding.G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_G723*

 - static readonly Guid [Acelp](#) = new Guid((short)AudioEncoding.Acelp & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
- WAVE_FORMAT_ACELP*

- static readonly Guid [RawAac](#) = new Guid((short)AudioEncoding.RawAac & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_RAW_AAC1
- static readonly Guid [WAVE_FORMAT_RHETOREX_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_RHETOREX_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_RHETOREX_ADPCM
- static readonly Guid [WAVE_FORMAT_IRAT](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_IRAT & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_IRAT
- static readonly Guid [WAVE_FORMAT_VIVO_G723](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VIVO_G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_VIVO_G723
- static readonly Guid [WAVE_FORMAT_VIVO_SIREN](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VIVO_SIREN & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_VIVO_SIREN
- static readonly Guid [WAVE_FORMAT_DIGITAL_G723](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_DIGITAL_G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DIGITAL_G723
- static readonly Guid [WAVE_FORMAT_SANYO_LD_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SANYO_LD_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SANYO_LD_ADPCM
- static readonly Guid [WAVE_FORMAT_SIPROLAB_ACEPLNET](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_ACEPLNET & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SIPROLAB_ACEPLNET
- static readonly Guid [WAVE_FORMAT_SIPROLAB_ACELP4800](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_ACELP4800 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SIPROLAB_ACELP4800
- static readonly Guid [WAVE_FORMAT_SIPROLAB_ACELP8V3](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_ACELP8V3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SIPROLAB_ACELP8V3
- static readonly Guid [WAVE_FORMAT_SIPROLAB_G729](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_G729 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SIPROLAB_G729
- static readonly Guid [WAVE_FORMAT_SIPROLAB_G729A](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_G729A & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SIPROLAB_G729A
- static readonly Guid [WAVE_FORMAT_SIPROLAB_KELVIN](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_KELVIN & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SIPROLAB_KELVIN
- static readonly Guid [WAVE_FORMAT_G726ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_G726ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_G726ADPCM
- static readonly Guid [WAVE_FORMAT_QUALCOMM_PUREVOICE](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_QUALCOMM_PUREVOICE & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_QUALCOMM_PUREVOICE

- static readonly Guid [WAVE_FORMAT_QUALCOMM_HALFRATE](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_QUALCOMM_HALFRATE & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_QUALCOMM_HALFRATE
- static readonly Guid [WAVE_FORMAT_TUBGSM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_TUBGSM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_TUBGSM
- static readonly Guid [WAVE_FORMAT_MSAUDIO1](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_MSAUDIO1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_MSAUDIO1
- static readonly Guid [WindowsMediaAudio](#) = new Guid((short)AudioEncoding.WindowsMediaAudio & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
Windows Media Audio, WAVE_FORMAT_WMAUDIO2, Microsoft Corporation
- static readonly Guid [WindowsMediaAudioProfessional](#) = new Guid((short)AudioEncoding.WindowsMediaAudioProfessional & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
Windows Media Audio Professional WAVE_FORMAT_WMAUDIO3, Microsoft Corporation
- static readonly Guid [WindowsMediaAudioLossless](#) = new Guid((short)AudioEncoding.WindowsMediaAudioLossless & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
Windows Media Audio Lossless, WAVE_FORMAT_WMAUDIO_LOSSLESS
- static readonly Guid [WindowsMediaAudioSpdif](#) = new Guid((short)AudioEncoding.WindowsMediaAudioSpdif & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
Windows Media Audio Professional over SPDIF WAVE_FORMAT_WMASPDIF (0x0164)
- static readonly Guid [WAVE_FORMAT_UNISYS_NAP_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_UNISYS_NAP_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_UNISYS_NAP_ADPCM
- static readonly Guid [WAVE_FORMAT_UNISYS_NAP_ULAW](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_UNISYS_NAP_ULAW & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_UNISYS_NAP_ULAW
- static readonly Guid [WAVE_FORMAT_UNISYS_NAP_ALAW](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_UNISYS_NAP_ALAW & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_UNISYS_NAP_ALAW
- static readonly Guid [WAVE_FORMAT_UNISYS_NAP_16K](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_UNISYS_NAP_16K & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_UNISYS_NAP_16K
- static readonly Guid [WAVE_FORMAT_CREATIVE_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_CREATIVE_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_CREATIVE_ADPCM
- static readonly Guid [WAVE_FORMAT_CREATIVE_FASTSPEECH8](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_CREATIVE_FASTSPEECH8 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_CREATIVE_FASTSPEECH8
- static readonly Guid [WAVE_FORMAT_CREATIVE_FASTSPEECH10](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_CREATIVE_FASTSPEECH10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_CREATIVE_FASTSPEECH10
- static readonly Guid [WAVE_FORMAT_UHER_ADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_UHER_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_UHER_ADPCM

- static readonly Guid [WAVE_FORMAT_QUARTERDECK](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_QUARTERDECK & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_QUARTERDECK
- static readonly Guid [WAVE_FORMAT_ILINK_VC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ILINK_VC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_ILINK_VC
- static readonly Guid [WAVE_FORMAT_RAW_SPORT](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_RAW_SPORT & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_RAW_SPORT
- static readonly Guid [WAVE_FORMAT_ESST_AC3](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_ESST_AC3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_ESST_AC3
- static readonly Guid [WAVE_FORMAT_IPI_HSX](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_IPI_HSX & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_IPI_HSX
- static readonly Guid [WAVE_FORMAT_IPI_RPELP](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_IPI_RPELP & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_IPI_RPELP
- static readonly Guid [WAVE_FORMAT_CS2](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_CS2 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_CS2
- static readonly Guid [WAVE_FORMAT_SONY_SCX](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SONY_SCX & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_SONY_SCX
- static readonly Guid [WAVE_FORMAT_FM_TOWNS_SND](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_FM_TOWNS_SND & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_FM_TOWNS_SND
- static readonly Guid [WAVE_FORMAT_BTV_DIGITAL](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_BTV_DIGITAL & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_BTV_DIGITAL
- static readonly Guid [WAVE_FORMAT_QDESIGN_MUSIC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_QDESIGN_MUSIC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_QDESIGN_MUSIC
- static readonly Guid [WAVE_FORMAT_VME_VMPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_VME_VMPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_VME_VMPCM
- static readonly Guid [WAVE_FORMAT_TPC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_TPC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_TPC
- static readonly Guid [WAVE_FORMAT_OLIGSM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_OLIGSM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_OLIGSM
- static readonly Guid [WAVE_FORMAT_OLIADPCM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_OLIADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_OLIADPCM
- static readonly Guid [WAVE_FORMAT_OLICELP](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_OLICELP & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_OLICELP
- static readonly Guid [WAVE_FORMAT_OLISBC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_OLISBC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_OLISBC
- static readonly Guid [WAVE_FORMAT_OLIOPR](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_OLIOPR & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_OLIOPR

- static readonly Guid [WAVE_FORMAT_LH_CODEC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_LH_CODEC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_LH_CODEC

- static readonly Guid [WAVE_FORMAT_NORRIS](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_NORRIS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_NORRIS

- static readonly Guid [WAVE_FORMAT_SOUNDSPACE_MUSICOMPRESS](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_SOUNDSPACE_MUSICOMPRESS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_SOUNDSPACE_MUSICOMPRESS

- static readonly Guid [MPEG_ADTS_AAC](#) = new Guid((short)AudioEncoding.MPEG_ADTS_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

Advanced Audio Coding (AAC) audio in Audio Data Transport Stream (ADTS) format. The format block is a WAVEFORMATX structure with wFormatTag equal to WAVE_FORMAT_MPEG_ADTS_AAC.

- static readonly Guid [MPEG_RAW_AAC](#) = new Guid((short)AudioEncoding.MPEG_RAW_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

MPEG_RAW_AAC

- static readonly Guid [MPEG_LOAS](#) = new Guid((short)AudioEncoding.MPEG_LOAS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

MPEG-4 audio transport stream with a synchronization layer (LOAS) and a multiplex layer (LATM). The format block is a WAVEFORMATX structure with wFormatTag equal to WAVE_FORMAT_MPEG_LOAS. See .

- static readonly Guid [NOKIA_MPEG_ADTS_AAC](#) = new Guid((short)AudioEncoding.NOKIA_MPEG_ADTS_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

NOKIA_MPEG_ADTS_AAC

- static readonly Guid [NOKIA_MPEG_RAW_AAC](#) = new Guid((short)AudioEncoding.NOKIA_MPEG_RAW_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

NOKIA_MPEG_RAW_AAC

- static readonly Guid [VODAFONE_MPEG_ADTS_AAC](#) = new Guid((short)AudioEncoding.VODAFONE_MPEG_ADTS_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

VODAFONE_MPEG_ADTS_AAC

- static readonly Guid [VODAFONE_MPEG_RAW_AAC](#) = new Guid((short)AudioEncoding.VODAFONE_MPEG_RAW_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

VODAFONE_MPEG_RAW_AAC

- static readonly Guid [MPEG_HEAAC](#) = new Guid((short)AudioEncoding.MPEG_HEAAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

High-Efficiency Advanced Audio Coding (HE-AAC) stream. The format block is an HEAACWAVEFORMAT structure. See .

- static readonly Guid [WAVE_FORMAT_DVM](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_DVM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_DVM

- static readonly Guid [Vorbis1](#) = new Guid((short)AudioEncoding.Vorbis1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_VORBIS1 "Og" Original stream compatible

- static readonly Guid [Vorbis2](#) = new Guid((short)AudioEncoding.Vorbis2 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_VORBIS2 "Pg" Have independent header

- static readonly Guid [Vorbis3](#) = new Guid((short)AudioEncoding.Vorbis3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_VORBIS3 "Qg" Have no codebook header

- static readonly Guid [Vorbis1P](#) = new Guid((short)AudioEncoding.Vorbis1P & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)

WAVE_FORMAT_VORBIS1P "og" Original stream compatible

- static readonly Guid [Vorbis2P](#) = new Guid((short)AudioEncoding.Vorbis2P & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_VORBIS2P "pg" Have independent headers
- static readonly Guid [Vorbis3P](#) = new Guid((short)AudioEncoding.Vorbis3P & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_VORBIS3P "qg" Have no codebook header
- static readonly Guid [WAVE_FORMAT_RAW_AAC1](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_↵
RAW_AAC1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
Raw AAC1
- static readonly Guid [WAVE_FORMAT_WMAVOICE9](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_↵
_WMAVOICE9 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
Windows Media Audio Voice (WMA Voice)
- static readonly Guid [Extensible](#) = new Guid((short)AudioEncoding.Extensible & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
Extensible
- static readonly Guid [WAVE_FORMAT_DEVELOPMENT](#) = new Guid((short)AudioEncoding.WAVE_FORM↵
AT_DEVELOPMENT & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
WAVE_FORMAT_DEVELOPMENT
- static readonly Guid [WAVE_FORMAT_FLAC](#) = new Guid((short)AudioEncoding.WAVE_FORMAT_FLAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
FLAC

4.8.1 Detailed Description

Defines [AudioSubTypes](#) and provides methods to convert between [AudioEncoding](#)-values and [AudioSubTypes](#)-values.

[AudioSubTypes](#) are used by the [WaveFormatExtensible](#), the [MFMediaType](#) and the [MediaType](#) class.

4.8.2 Member Function Documentation

4.8.2.1 EncodingFromSubType()

```
static AudioEncoding EncodingFromSubType (
    Guid audioSubType ) [static]
```

Converts a [AudioSubTypes](#)-value to a [AudioEncoding](#)-value.

Parameters

<i>audioSubType</i>	The AudioSubTypes -value to convert to the equivalent AudioEncoding -value.
---------------------	---

Returns

The [AudioEncoding](#) which belongs to the specified *audioSubType* .

4.8.2.2 SubTypeFromEncoding()

```
static Guid SubTypeFromEncoding (
    AudioEncoding audioEncoding ) [static]
```

Converts a [AudioEncoding](#) value to a [AudioSubTypes](#)-value.

Parameters

<i>audioEncoding</i>	The AudioEncoding to convert to the equivalent AudioSubTypes -value.
----------------------	--

Returns

The [AudioSubTypes](#)-value which belongs to the specified *audioEncoding* .

4.8.3 Member Data Documentation

4.8.3.1 Acelp

```
readonly Guid Acelp = new Guid((short)AudioEncoding.Acelp & 0x0000FFFF, 0x0000, 0x0010, 0x80,
0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ACELP

4.8.3.2 Adpcm

```
readonly Guid Adpcm = new Guid((short)AudioEncoding.Adpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80,
0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ADPCM Microsoft Corporation

4.8.3.3 ALaw

```
readonly Guid ALaw = new Guid((short)AudioEncoding.ALaw & 0x0000FFFF, 0x0000, 0x0010, 0x80,
0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ALAW Microsoft Corporation

4.8.3.4 AntexAdpcme

```
readonly Guid AntexAdpcme = new Guid((short)AudioEncoding.AntexAdpcme & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ANTEX_ADPCME, Antex Electronics Corporation

4.8.3.5 Aptx

```
readonly Guid Aptx = new Guid((short)AudioEncoding.Aptx & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_APTX Audio Processing Technology

4.8.3.6 AudioFileAf10

```
readonly Guid AudioFileAf10 = new Guid((short)AudioEncoding.AudioFileAf10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_AUDIOFILE_AF10, Virtual Music, Inc.

4.8.3.7 AudioFileAf36

```
readonly Guid AudioFileAf36 = new Guid((short)AudioEncoding.AudioFileAf36 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_AUDIOFILE_AF36, Virtual Music, Inc.

4.8.3.8 ControlResCr10

```
readonly Guid ControlResCr10 = new Guid((short)AudioEncoding.ControlResCr10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CONTROL_RES_CR10, Control Resources Limited

4.8.3.9 ControlResVqlpc

```
readonly Guid ControlResVqlpc = new Guid((short)AudioEncoding.ControlResVqlpc & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CONTROL_RES_VQLPC, Control Resources Limited

4.8.3.10 CUCodec

```
readonly Guid CUCodec = new Guid((short)AudioEncoding.CUCodec & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CU_CODECS Hewlett-Packard Company

4.8.3.11 DialogicOkiAdpcm

```
readonly Guid DialogicOkiAdpcm = new Guid((short)AudioEncoding.DialogicOkiAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DIALOGIC_OKI_ADPCM Dialogic Corporation

4.8.3.12 DigiAdpcm

```
readonly Guid DigiAdpcm = new Guid((short)AudioEncoding.DigiAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DIGIADPCM, DSP Solutions, Inc.

4.8.3.13 DigiFix

```
readonly Guid DigiFix = new Guid((short)AudioEncoding.DigiFix & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DIGIFIX DSP Solutions, Inc.

4.8.3.14 DigiReal

```
readonly Guid DigiReal = new Guid((short)AudioEncoding.DigiReal & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DIGIREAL, DSP Solutions, Inc.

4.8.3.15 DigiStd

```
readonly Guid DigiStd = new Guid((short)AudioEncoding.DigiStd & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DIGISTD DSP Solutions, Inc.

4.8.3.16 DolbyAc2

```
readonly Guid DolbyAc2 = new Guid((short)AudioEncoding.DolbyAc2 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DOLBY_AC2, Dolby Laboratories

4.8.3.17 Drm

```
readonly Guid Drm = new Guid((short)AudioEncoding.Drm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DRM Microsoft Corporation

4.8.3.18 DspGroupTrueSpeech

```
readonly Guid DspGroupTrueSpeech = new Guid((short)AudioEncoding.DspGroupTrueSpeech & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DSPGROUP_TRUESPEECH DSP Group, Inc

4.8.3.19 Dts

```
readonly Guid Dts = new Guid((short)AudioEncoding.Dts & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DTS Microsoft Corporation

4.8.3.20 DviAdpcm

```
readonly Guid DviAdpcm = new Guid((short)AudioEncoding.DviAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DVI_ADPCM Intel Corporation

4.8.3.21 EchoSpeechCorporation1

```
readonly Guid EchoSpeechCorporation1 = new Guid((short)AudioEncoding.EchoSpeechCorporation1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ECHOSC1 Echo Speech Corporation

4.8.3.22 Extensible

```
readonly Guid Extensible = new Guid((short)AudioEncoding.Extensible & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Extensible

4.8.3.23 G723

```
readonly Guid G723 = new Guid((short)AudioEncoding.G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G723

4.8.3.24 G723Adpcm

```
readonly Guid G723Adpcm = new Guid((short)AudioEncoding.G723Adpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G723_ADPCM Antex Electronics Corporation

4.8.3.25 G729

```
readonly Guid G729 = new Guid((short)AudioEncoding.G729 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G729

4.8.3.26 Gsm

```
readonly Guid Gsm = new Guid((short)AudioEncoding.Gsm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_GSM

4.8.3.27 Gsm610

```
readonly Guid Gsm610 = new Guid((short)AudioEncoding.Gsm610 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_GSM610, Microsoft Corporation

4.8.3.28 IbmCvsd

```
readonly Guid IbmCvsd = new Guid((short)AudioEncoding.IbmCvsd & 0x0000FFFF, 0x0000, 0x0010,  
0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_IBM_CVSD IBM Corporation

4.8.3.29 IeeeFloat

```
readonly Guid IeeeFloat = new Guid((short)AudioEncoding.IeeeFloat & 0x0000FFFF, 0x0000, 0x0010,  
0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_IEEE_FLOAT Microsoft Corporation

4.8.3.30 ImaAdpcm

```
readonly Guid ImaAdpcm = new Guid((short)AudioEncoding.ImaAdpcm & 0x0000FFFF, 0x0000, 0x0010,  
0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_IMA_ADPCM Intel Corporation

4.8.3.31 Lrc

```
readonly Guid Lrc = new Guid((short)AudioEncoding.Lrc & 0x0000FFFF, 0x0000, 0x0010, 0x80,  
0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_LRC, Merging Technologies S.A.

4.8.3.32 MediaspaceAdpcm

```
readonly Guid MediaspaceAdpcm = new Guid((short)AudioEncoding.MediaspaceAdpcm & 0x0000FFFF,  
0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MEDIASPACE_ADPCM Videologic

4.8.3.33 MediaTypeAudio

```
readonly Guid MediaTypeAudio = new Guid("73647561-0000-0010-8000-00AA00389B71") [static]
```

The Major Type for Audio media types.

4.8.3.34 MediaVisionAdpcm

```
readonly Guid MediaVisionAdpcm = new Guid((short)AudioEncoding.MediaVisionAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MEDIAVISION_ADPCM Media Vision, Inc.

4.8.3.35 Mpeg

```
readonly Guid Mpeg = new Guid((short)AudioEncoding.Mpeg & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MPEG, Microsoft Corporation

4.8.3.36 MPEG_ADTS_AAC

```
readonly Guid MPEG_ADTS_AAC = new Guid((short)AudioEncoding.MPEG_ADTS_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Advanced Audio Coding (AAC) audio in Audio Data Transport Stream (ADTS) format. The format block is a WAV↔EFORMATEX structure with wFormatTag equal to WAVE_FORMAT_MPEG_ADTS_AAC.

The WAVEFORMATEX structure specifies the core AAC-LC sample rate and number of channels, prior to applying spectral band replication (SBR) or parametric stereo (PS) tools, if present. No additional data is required after the WAVEFORMATEX structure.

<http://msdn.microsoft.com/en-us/library/dd317599%28VS.85%29.aspx>

4.8.3.37 MPEG_HEAAC

```
readonly Guid MPEG_HEAAC = new Guid((short)AudioEncoding.MPEG_HEAAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

High-Efficiency Advanced Audio Coding (HE-AAC) stream. The format block is an HEAACWAVEFORMAT structure. See .

4.8.3.38 MPEG_LOAS

```
readonly Guid MPEG_LOAS = new Guid((short)AudioEncoding.MPEG_LOAS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

MPEG-4 audio transport stream with a synchronization layer (LOAS) and a multiplex layer (LATM). The format block is a WAVEFORMATEX structure with wFormatTag equal to WAVE_FORMAT_MPEG_LOAS. See .

The WAVEFORMATEX structure specifies the core AAC-LC sample rate and number of channels, prior to applying spectral SBR or PS tools, if present. No additional data is required after the WAVEFORMATEX structure.

4.8.3.39 MPEG_RAW_AAC

```
readonly Guid MPEG_RAW_AAC = new Guid((short)AudioEncoding.MPEG_RAW_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

MPEG_RAW_AAC

Source wmCodec.h

4.8.3.40 MpegLayer3

```
readonly Guid MpegLayer3 = new Guid((short)AudioEncoding.MpegLayer3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MPEGLAYER3, ISO/MPEG Layer3 Format Tag

4.8.3.41 MsnAudio

```
readonly Guid MsnAudio = new Guid((short)AudioEncoding.MsnAudio & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MSNAUDIO, Microsoft Corporation

4.8.3.42 MuLaw

```
readonly Guid MuLaw = new Guid((short)AudioEncoding.MuLaw & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MULAW Microsoft Corporation

4.8.3.43 NOKIA_MPEG_ADTS_AAC

```
readonly Guid NOKIA_MPEG_ADTS_AAC = new Guid((short)AudioEncoding.NOKIA_MPEG_ADTS_AAC & 0x0000←FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

NOKIA_MPEG_ADTS_AAC

Source wmCodec.h

4.8.3.44 NOKIA_MPEG_RAW_AAC

```
readonly Guid NOKIA_MPEG_RAW_AAC = new Guid((short)AudioEncoding.NOKIA_MPEG_RAW_AAC & 0x0000←FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

NOKIA_MPEG_RAW_AAC

Source wmCodec.h

4.8.3.45 OkiAdpcm

```
readonly Guid OkiAdpcm = new Guid((short)AudioEncoding.OkiAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_OKI_ADPCM OKI

4.8.3.46 Pcm

```
readonly Guid Pcm = new Guid((short)AudioEncoding.Pcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_PCM Microsoft Corporation

4.8.3.47 Prosody1612

```
readonly Guid Prosody1612 = new Guid((short)AudioEncoding.Prosody1612 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_PROSODY_1612, Aculab plc

4.8.3.48 RawAac

```
readonly Guid RawAac = new Guid((short)AudioEncoding.RawAac & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_RAW_AAC1

4.8.3.49 SierraAdpcm

```
readonly Guid SierraAdpcm = new Guid((short)AudioEncoding.SierraAdpcm & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SIERRA_ADPCM Sierra Semiconductor Corp

4.8.3.50 SonarC

```
readonly Guid SonarC = new Guid((short)AudioEncoding.SonarC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SONARC Speech Compression

4.8.3.51 Unknown

```
readonly Guid Unknown = new Guid((short)AudioEncoding.Unknown & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_UNKNOWN, Microsoft Corporation

4.8.3.52 VODAFONE_MPEG_ADTS_AAC

```
readonly Guid VODAFONE_MPEG_ADTS_AAC = new Guid((short)AudioEncoding.VODAFONE_MPEG_ADTS_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

VODAFONE_MPEG_ADTS_AAC

Source wmCodec.h

4.8.3.53 VODAFONE_MPEG_RAW_AAC

```
readonly Guid VODAFONE_MPEG_RAW_AAC = new Guid((short)AudioEncoding.VODAFONE_MPEG_RAW_AAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

VODAFONE_MPEG_RAW_AAC

Source wmCodec.h

4.8.3.54 Vorbis1

```
readonly Guid Vorbis1 = new Guid((short)AudioEncoding.Vorbis1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VORBIS1 "Og" Original stream compatible

4.8.3.55 Vorbis1P

```
readonly Guid Vorbis1P = new Guid((short)AudioEncoding.Vorbis1P & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VORBIS1P "og" Original stream compatible

4.8.3.56 Vorbis2

```
readonly Guid Vorbis2 = new Guid((short)AudioEncoding.Vorbis2 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VORBIS2 "Pg" Have independent header

4.8.3.57 Vorbis2P

```
readonly Guid Vorbis2P = new Guid((short)AudioEncoding.Vorbis2P & 0x0000FFFF, 0x0000, 0x0010,
0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VORBIS2P "pg" Have independent headere

4.8.3.58 Vorbis3

```
readonly Guid Vorbis3 = new Guid((short)AudioEncoding.Vorbis3 & 0x0000FFFF, 0x0000, 0x0010,
0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VORBIS3 "Qg" Have no codebook header

4.8.3.59 Vorbis3P

```
readonly Guid Vorbis3P = new Guid((short)AudioEncoding.Vorbis3P & 0x0000FFFF, 0x0000, 0x0010,
0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VORBIS3P "qg" Have no codebook header

4.8.3.60 Vselp

```
readonly Guid Vselp = new Guid((short)AudioEncoding.Vselp & 0x0000FFFF, 0x0000, 0x0010, 0x80,
0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VSELP Compaq Computer Corp.

4.8.3.61 WAVE_FORMAT_BTV_DIGITAL

```
readonly Guid WAVE_FORMAT_BTV_DIGITAL = new Guid((short)AudioEncoding.WAVE_FORMAT_BTV_DIGITAL
& 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_BTV_DIGITAL

4.8.3.62 WAVE_FORMAT_CANOPUS_ATRAC

```
readonly Guid WAVE_FORMAT_CANOPUS_ATRAC = new Guid((short)AudioEncoding.WAVE_FORMAT_CANOPUS_↵
ATRAC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CANOPUS_ATRAC

4.8.3.63 WAVE_FORMAT_CIRRUS

```
readonly Guid WAVE_FORMAT_CIRRUS = new Guid((short)AudioEncoding.WAVE_FORMAT_CIRRUS & 0x0000←  
FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CIRRUS

4.8.3.64 WAVE_FORMAT_CREATIVE_ADPCM

```
readonly Guid WAVE_FORMAT_CREATIVE_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_CREATIV←  
E_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CREATIVE_ADPCM

4.8.3.65 WAVE_FORMAT_CREATIVE_FASTSPEECH10

```
readonly Guid WAVE_FORMAT_CREATIVE_FASTSPEECH10 = new Guid((short)AudioEncoding.WAVE_FORMAT_←  
CREATIVE_FASTSPEECH10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b,  
0x71) [static]
```

WAVE_FORMAT_CREATIVE_FASTSPEECH10

4.8.3.66 WAVE_FORMAT_CREATIVE_FASTSPEECH8

```
readonly Guid WAVE_FORMAT_CREATIVE_FASTSPEECH8 = new Guid((short)AudioEncoding.WAVE_FORMAT_←  
CREATIVE_FASTSPEECH8 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b,  
0x71) [static]
```

WAVE_FORMAT_CREATIVE_FASTSPEECH8

4.8.3.67 WAVE_FORMAT_CS2

```
readonly Guid WAVE_FORMAT_CS2 = new Guid((short)AudioEncoding.WAVE_FORMAT_CS2 & 0x0000FFFF,  
0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CS2

4.8.3.68 WAVE_FORMAT_CS_IMAADPCM

```
readonly Guid WAVE_FORMAT_CS_IMAADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_CS_IMAADPCM
& 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_CS_IMAADPCM

4.8.3.69 WAVE_FORMAT_DEVELOPMENT

```
readonly Guid WAVE_FORMAT_DEVELOPMENT = new Guid((short)AudioEncoding.WAVE_FORMAT_DEVELOPMENT
& 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DEVELOPMENT

4.8.3.70 WAVE_FORMAT_DF_G726

```
readonly Guid WAVE_FORMAT_DF_G726 = new Guid((short)AudioEncoding.WAVE_FORMAT_DF_G726 & 0x0000←
FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DF_G726

4.8.3.71 WAVE_FORMAT_DF_GSM610

```
readonly Guid WAVE_FORMAT_DF_GSM610 = new Guid((short)AudioEncoding.WAVE_FORMAT_DF_GSM610 &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DF_GSM610

4.8.3.72 WAVE_FORMAT_DIGITAL_G723

```
readonly Guid WAVE_FORMAT_DIGITAL_G723 = new Guid((short)AudioEncoding.WAVE_FORMAT_DIGITAL_←
G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DIGITAL_G723

4.8.3.73 WAVE_FORMAT_DOLBY_AC3_SPDIF

```
readonly Guid WAVE_FORMAT_DOLBY_AC3_SPDIF = new Guid((short)AudioEncoding.WAVE_FORMAT_DOLBY_AC3_SPDIF & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DOLBY_AC3_SPDIF

4.8.3.74 WAVE_FORMAT_DSAT_DISPLAY

```
readonly Guid WAVE_FORMAT_DSAT_DISPLAY = new Guid((short)AudioEncoding.WAVE_FORMAT_DSAT_DISPLAY & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DSAT_DISPLAY

4.8.3.75 WAVE_FORMAT_DVM

```
readonly Guid WAVE_FORMAT_DVM = new Guid((short)AudioEncoding.WAVE_FORMAT_DVM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_DVM

4.8.3.76 WAVE_FORMAT_ECHOSC3

```
readonly Guid WAVE_FORMAT_ECHOSC3 = new Guid((short)AudioEncoding.WAVE_FORMAT_ECHOSC3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ECHOSC3

4.8.3.77 WAVE_FORMAT_ESPCM

```
readonly Guid WAVE_FORMAT_ESPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_ESPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ESPCM

4.8.3.78 WAVE_FORMAT_ESST_AC3

```
readonly Guid WAVE_FORMAT_ESST_AC3 = new Guid((short)AudioEncoding.WAVE_FORMAT_ESST_AC3 &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ESST_AC3

4.8.3.79 WAVE_FORMAT_FLAC

```
readonly Guid WAVE_FORMAT_FLAC = new Guid((short)AudioEncoding.WAVE_FORMAT_FLAC & 0x0000FFFF,
0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

FLAC

4.8.3.80 WAVE_FORMAT_FM_TOWNS_SND

```
readonly Guid WAVE_FORMAT_FM_TOWNS_SND = new Guid((short)AudioEncoding.WAVE_FORMAT_FM_TOWNS_↵
SND & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_FM_TOWNS_SND

4.8.3.81 WAVE_FORMAT_G721_ADPCM

```
readonly Guid WAVE_FORMAT_G721_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_G721_ADPCM &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G721_ADPCM

4.8.3.82 WAVE_FORMAT_G722_ADPCM

```
readonly Guid WAVE_FORMAT_G722_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_G722_ADPCM &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G722_ADPCM

4.8.3.83 WAVE_FORMAT_G726_ADPCM

```
readonly Guid WAVE_FORMAT_G726_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_G726_ADPCM &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G726_ADPCM

4.8.3.84 WAVE_FORMAT_G726ADPCM

```
readonly Guid WAVE_FORMAT_G726ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_G726ADPCM &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G726ADPCM

4.8.3.85 WAVE_FORMAT_G728_CELP

```
readonly Guid WAVE_FORMAT_G728_CELP = new Guid((short)AudioEncoding.WAVE_FORMAT_G728_CELP &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G728_CELP

4.8.3.86 WAVE_FORMAT_G729A

```
readonly Guid WAVE_FORMAT_G729A = new Guid((short)AudioEncoding.WAVE_FORMAT_G729A & 0x0000FF↵
FF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_G729A

4.8.3.87 WAVE_FORMAT_ILINK_VC

```
readonly Guid WAVE_FORMAT_ILINK_VC = new Guid((short)AudioEncoding.WAVE_FORMAT_ILINK_VC &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ILINK_VC

4.8.3.88 WAVE_FORMAT_IPI_HSX

```
readonly Guid WAVE_FORMAT_IPI_HSX = new Guid((short)AudioEncoding.WAVE_FORMAT_IPI_HSX & 0x0000↵
FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_IPI_HSX

4.8.3.89 WAVE_FORMAT_IPI_RPELP

```
readonly Guid WAVE_FORMAT_IPI_RPELP = new Guid((short)AudioEncoding.WAVE_FORMAT_IPI_RPELP &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_IPI_RPELP

4.8.3.90 WAVE_FORMAT_IRAT

```
readonly Guid WAVE_FORMAT_IRAT = new Guid((short)AudioEncoding.WAVE_FORMAT_IRAT & 0x0000FFFF,
0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_IRAT

4.8.3.91 WAVE_FORMAT_ISIAUDIO

```
readonly Guid WAVE_FORMAT_ISIAUDIO = new Guid((short)AudioEncoding.WAVE_FORMAT_ISIAUDIO &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ISIAUDIO

4.8.3.92 WAVE_FORMAT_LH_CODEC

```
readonly Guid WAVE_FORMAT_LH_CODEC = new Guid((short)AudioEncoding.WAVE_FORMAT_LH_CODEC &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_LH_CODEC

4.8.3.93 WAVE_FORMAT_LUCENT_G723

```
readonly Guid WAVE_FORMAT_LUCENT_G723 = new Guid((short)AudioEncoding.WAVE_FORMAT_LUCENT_G723
& 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_LUCENT_G723

4.8.3.94 WAVE_FORMAT_MALDEN_PHONYTALK

```
readonly Guid WAVE_FORMAT_MALDEN_PHONYTALK = new Guid((short)AudioEncoding.WAVE_FORMAT_MALDEN_PHONYTALK & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)
[static]
```

WAVE_FORMAT_MALDEN_PHONYTALK

4.8.3.95 WAVE_FORMAT_MEDIASONIC_G723

```
readonly Guid WAVE_FORMAT_MEDIASONIC_G723 = new Guid((short)AudioEncoding.WAVE_FORMAT_MEDIASONIC_G723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MEDIASONIC_G723

4.8.3.96 WAVE_FORMAT_MSAUDIO1

```
readonly Guid WAVE_FORMAT_MSAUDIO1 = new Guid((short)AudioEncoding.WAVE_FORMAT_MSAUDIO1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MSAUDIO1

4.8.3.97 WAVE_FORMAT_MSG723

```
readonly Guid WAVE_FORMAT_MSG723 = new Guid((short)AudioEncoding.WAVE_FORMAT_MSG723 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MSG723

4.8.3.98 WAVE_FORMAT_MSRT24

```
readonly Guid WAVE_FORMAT_MSRT24 = new Guid((short)AudioEncoding.WAVE_FORMAT_MSRT24 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MSRT24

4.8.3.99 WAVE_FORMAT_MVI_MVI2

```
readonly Guid WAVE_FORMAT_MVI_MVI2 = new Guid((short)AudioEncoding.WAVE_FORMAT_MVI_MVI2 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_MVI_MVI2

4.8.3.100 WAVE_FORMAT_NMS_VBXADPCM

```
readonly Guid WAVE_FORMAT_NMS_VBXADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_NMS_VBXADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_NMS_VBXADPCM

4.8.3.101 WAVE_FORMAT_NORRIS

```
readonly Guid WAVE_FORMAT_NORRIS = new Guid((short)AudioEncoding.WAVE_FORMAT_NORRIS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_NORRIS

4.8.3.102 WAVE_FORMAT_OLIADPCM

```
readonly Guid WAVE_FORMAT_OLIADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_OLIADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_OLIADPCM

4.8.3.103 WAVE_FORMAT_OLICELP

```
readonly Guid WAVE_FORMAT_OLICELP = new Guid((short)AudioEncoding.WAVE_FORMAT_OLICELP & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_OLICELP

4.8.3.104 WAVE_FORMAT_OLIGSM

```
readonly Guid WAVE_FORMAT_OLIGSM = new Guid((short)AudioEncoding.WAVE_FORMAT_OLIGSM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_OLIGSM

4.8.3.105 WAVE_FORMAT_OLIOPR

```
readonly Guid WAVE_FORMAT_OLIOPR = new Guid((short)AudioEncoding.WAVE_FORMAT_OLIOPR & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_OLIOPR

4.8.3.106 WAVE_FORMAT_OLISBC

```
readonly Guid WAVE_FORMAT_OLISBC = new Guid((short)AudioEncoding.WAVE_FORMAT_OLISBC & 0x0000←  
FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_OLISBC

4.8.3.107 WAVE_FORMAT_ONLIVE

```
readonly Guid WAVE_FORMAT_ONLIVE = new Guid((short)AudioEncoding.WAVE_FORMAT_ONLIVE & 0x0000←  
FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ONLIVE

4.8.3.108 WAVE_FORMAT_PAC

```
readonly Guid WAVE_FORMAT_PAC = new Guid((short)AudioEncoding.WAVE_FORMAT_PAC & 0x0000FFFF,  
0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_PAC

4.8.3.109 WAVE_FORMAT_PACKED

```
readonly Guid WAVE_FORMAT_PACKED = new Guid((short)AudioEncoding.WAVE_FORMAT_PACKED & 0x0000←  
FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_PACKED

4.8.3.110 WAVE_FORMAT_PHILIPS_LPCBB

```
readonly Guid WAVE_FORMAT_PHILIPS_LPCBB = new Guid((short)AudioEncoding.WAVE_FORMAT_PHILIPS_←  
LPCBB & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_PHILIPS_LPCBB

4.8.3.111 WAVE_FORMAT_PROSODY_8KBPS

```
readonly Guid WAVE_FORMAT_PROSODY_8KBPS = new Guid((short)AudioEncoding.WAVE_FORMAT_PROSODY_←  
8KBPS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_PROSODY_8KBPS

4.8.3.112 WAVE_FORMAT_QDESIGN_MUSIC

```
readonly Guid WAVE_FORMAT_QDESIGN_MUSIC = new Guid((short)AudioEncoding.WAVE_FORMAT_QDESIGN_MUSIC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_QDESIGN_MUSIC

4.8.3.113 WAVE_FORMAT_QUALCOMM_HALFRATE

```
readonly Guid WAVE_FORMAT_QUALCOMM_HALFRATE = new Guid((short)AudioEncoding.WAVE_FORMAT_QUALCOMM_HALFRATE & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_QUALCOMM_HALFRATE

4.8.3.114 WAVE_FORMAT_QUALCOMM_PUREVOICE

```
readonly Guid WAVE_FORMAT_QUALCOMM_PUREVOICE = new Guid((short)AudioEncoding.WAVE_FORMAT_QUALCOMM_PUREVOICE & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_QUALCOMM_PUREVOICE

4.8.3.115 WAVE_FORMAT_QUARTERDECK

```
readonly Guid WAVE_FORMAT_QUARTERDECK = new Guid((short)AudioEncoding.WAVE_FORMAT_QUARTERDECK & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_QUARTERDECK

4.8.3.116 WAVE_FORMAT_RAW_AAC1

```
readonly Guid WAVE_FORMAT_RAW_AAC1 = new Guid((short)AudioEncoding.WAVE_FORMAT_RAW_AAC1 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Raw AAC1

4.8.3.117 WAVE_FORMAT_RAW_SPORT

```
readonly Guid WAVE_FORMAT_RAW_SPORT = new Guid((short)AudioEncoding.WAVE_FORMAT_RAW_SPORT &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_RAW_SPORT

4.8.3.118 WAVE_FORMAT_RHETOREX_ADPCM

```
readonly Guid WAVE_FORMAT_RHETOREX_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_RHETOREX_ADPCM &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_RHETOREX_ADPCM

4.8.3.119 WAVE_FORMAT_ROCKWELL_ADPCM

```
readonly Guid WAVE_FORMAT_ROCKWELL_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_ROCKWELL_ADPCM &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ROCKWELL_ADPCM

4.8.3.120 WAVE_FORMAT_ROCKWELL_DIGITALK

```
readonly Guid WAVE_FORMAT_ROCKWELL_DIGITALK = new Guid((short)AudioEncoding.WAVE_FORMAT_ROCKWELL_DIGITALK &
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ROCKWELL_DIGITALK

4.8.3.121 WAVE_FORMAT_RT24

```
readonly Guid WAVE_FORMAT_RT24 = new Guid((short)AudioEncoding.WAVE_FORMAT_RT24 & 0x0000FFFF,
0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_RT24

4.8.3.122 WAVE_FORMAT_SANYO_LD_ADPCM

```
readonly Guid WAVE_FORMAT_SANYO_LD_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_SANYO_LD_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SANYO_LD_ADPCM

4.8.3.123 WAVE_FORMAT_SBC24

```
readonly Guid WAVE_FORMAT_SBC24 = new Guid((short)AudioEncoding.WAVE_FORMAT_SBC24 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SBC24

4.8.3.124 WAVE_FORMAT_SIPROLAB_ACELP4800

```
readonly Guid WAVE_FORMAT_SIPROLAB_ACELP4800 = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_ACELP4800 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SIPROLAB_ACELP4800

4.8.3.125 WAVE_FORMAT_SIPROLAB_ACELP8V3

```
readonly Guid WAVE_FORMAT_SIPROLAB_ACELP8V3 = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_ACELP8V3 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SIPROLAB_ACELP8V3

4.8.3.126 WAVE_FORMAT_SIPROLAB_ACEPLNET

```
readonly Guid WAVE_FORMAT_SIPROLAB_ACEPLNET = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB_ACEPLNET & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SIPROLAB_ACEPLNET

4.8.3.127 WAVE_FORMAT_SIPROLAB_G729

```
readonly Guid WAVE_FORMAT_SIPROLAB_G729 = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLAB↵  
_G729 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SIPROLAB_G729

4.8.3.128 WAVE_FORMAT_SIPROLAB_G729A

```
readonly Guid WAVE_FORMAT_SIPROLAB_G729A = new Guid((short)AudioEncoding.WAVE_FORMAT_SIPROLA↵  
B_G729A & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SIPROLAB_G729A

4.8.3.129 WAVE_FORMAT_SIPROLAB_KELVIN

```
readonly Guid WAVE_FORMAT_SIPROLAB_KELVIN = new Guid((short)AudioEncoding.WAVE_FORMAT_SIP↵  
ROLAB_KELVIN & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)  
[static]
```

WAVE_FORMAT_SIPROLAB_KELVIN

4.8.3.130 WAVE_FORMAT_SOFTSOUND

```
readonly Guid WAVE_FORMAT_SOFTSOUND = new Guid((short)AudioEncoding.WAVE_FORMAT_SOFTSOUND &  
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SOFTSOUND

4.8.3.131 WAVE_FORMAT_SONY_SCX

```
readonly Guid WAVE_FORMAT_SONY_SCX = new Guid((short)AudioEncoding.WAVE_FORMAT_SONY_SCX &  
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SONY_SCX

4.8.3.132 WAVE_FORMAT_SOUNDSPACE_MUSICOMPRESS

```
readonly Guid WAVE_FORMAT_SOUNDSPACE_MUSICOMPRESS = new Guid((short)AudioEncoding.WAVE_FORMAT_SOUNDSPACE_MUSICOMPRESS & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_SOUNDSPACE_MUSICOMPRESS

4.8.3.133 WAVE_FORMAT_TPC

```
readonly Guid WAVE_FORMAT_TPC = new Guid((short)AudioEncoding.WAVE_FORMAT_TPC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_TPC

4.8.3.134 WAVE_FORMAT_TUBGSM

```
readonly Guid WAVE_FORMAT_TUBGSM = new Guid((short)AudioEncoding.WAVE_FORMAT_TUBGSM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_TUBGSM

4.8.3.135 WAVE_FORMAT_UHER_ADPCM

```
readonly Guid WAVE_FORMAT_UHER_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_UHER_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_UHER_ADPCM

4.8.3.136 WAVE_FORMAT_UNISYS_NAP_16K

```
readonly Guid WAVE_FORMAT_UNISYS_NAP_16K = new Guid((short)AudioEncoding.WAVE_FORMAT_UNISYS_NAP_16K & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_UNISYS_NAP_16K

4.8.3.137 WAVE_FORMAT_UNISYS_NAP_ADPCM

```
readonly Guid WAVE_FORMAT_UNISYS_NAP_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_UNI↔  
SYS_NAP_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)  
[static]
```

WAVE_FORMAT_UNISYS_NAP_ADPCM

4.8.3.138 WAVE_FORMAT_UNISYS_NAP_ALAW

```
readonly Guid WAVE_FORMAT_UNISYS_NAP_ALAW = new Guid((short)AudioEncoding.WAVE_FORMAT_UNI↔  
SYS_NAP_ALAW & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)  
[static]
```

WAVE_FORMAT_UNISYS_NAP_ALAW

4.8.3.139 WAVE_FORMAT_UNISYS_NAP_ULAW

```
readonly Guid WAVE_FORMAT_UNISYS_NAP_ULAW = new Guid((short)AudioEncoding.WAVE_FORMAT_UNI↔  
SYS_NAP_ULAW & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71)  
[static]
```

WAVE_FORMAT_UNISYS_NAP_ULAW

4.8.3.140 WAVE_FORMAT_VIVO_G723

```
readonly Guid WAVE_FORMAT_VIVO_G723 = new Guid((short)AudioEncoding.WAVE_FORMAT_VIVO_G723 &  
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VIVO_G723

4.8.3.141 WAVE_FORMAT_VIVO_SIREN

```
readonly Guid WAVE_FORMAT_VIVO_SIREN = new Guid((short)AudioEncoding.WAVE_FORMAT_VIVO_SIREN &  
0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VIVO_SIREN

4.8.3.142 WAVE_FORMAT_VME_VMPCM

```
readonly Guid WAVE_FORMAT_VME_VMPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_VME_VMPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VME_VMPCM

4.8.3.143 WAVE_FORMAT_VOXWARE

```
readonly Guid WAVE_FORMAT_VOXWARE = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE

4.8.3.144 WAVE_FORMAT_VOXWARE_AC10

```
readonly Guid WAVE_FORMAT_VOXWARE_AC10 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC10 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_AC10

4.8.3.145 WAVE_FORMAT_VOXWARE_AC16

```
readonly Guid WAVE_FORMAT_VOXWARE_AC16 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC16 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_AC16

4.8.3.146 WAVE_FORMAT_VOXWARE_AC20

```
readonly Guid WAVE_FORMAT_VOXWARE_AC20 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC20 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_AC20

4.8.3.147 WAVE_FORMAT_VOXWARE_AC8

```
readonly Guid WAVE_FORMAT_VOXWARE_AC8 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_AC8 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_AC8

4.8.3.148 WAVE_FORMAT_VOXWARE_BYTE_ALIGNED

```
readonly Guid WAVE_FORMAT_VOXWARE_BYTE_ALIGNED = new Guid((short)AudioEncoding.WAVE_FORMAT_↵  
VOXWARE_BYTE_ALIGNED & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b,  
0x71) [static]
```

WAVE_FORMAT_VOXWARE_BYTE_ALIGNED

4.8.3.149 WAVE_FORMAT_VOXWARE_RT24

```
readonly Guid WAVE_FORMAT_VOXWARE_RT24 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_R↵  
T24 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_RT24

4.8.3.150 WAVE_FORMAT_VOXWARE_RT29

```
readonly Guid WAVE_FORMAT_VOXWARE_RT29 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_R↵  
T29 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_RT29

4.8.3.151 WAVE_FORMAT_VOXWARE_RT29HW

```
readonly Guid WAVE_FORMAT_VOXWARE_RT29HW = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE↵  
_RT29HW & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_RT29HW

4.8.3.152 WAVE_FORMAT_VOXWARE_TQ40

```
readonly Guid WAVE_FORMAT_VOXWARE_TQ40 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_T↵  
Q40 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_TQ40

4.8.3.153 WAVE_FORMAT_VOXWARE_TQ60

```
readonly Guid WAVE_FORMAT_VOXWARE_TQ60 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_TQ60 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_TQ60

4.8.3.154 WAVE_FORMAT_VOXWARE_VR12

```
readonly Guid WAVE_FORMAT_VOXWARE_VR12 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_VR12 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_VR12

4.8.3.155 WAVE_FORMAT_VOXWARE_VR18

```
readonly Guid WAVE_FORMAT_VOXWARE_VR18 = new Guid((short)AudioEncoding.WAVE_FORMAT_VOXWARE_VR18 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_VOXWARE_VR18

4.8.3.156 WAVE_FORMAT_WMAVOICE9

```
readonly Guid WAVE_FORMAT_WMAVOICE9 = new Guid((short)AudioEncoding.WAVE_FORMAT_WMAVOICE9 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Windows Media Audio Voice (WMA Voice)

4.8.3.157 WAVE_FORMAT_XEBEC

```
readonly Guid WAVE_FORMAT_XEBEC = new Guid((short)AudioEncoding.WAVE_FORMAT_XEBEC & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_XEBEC

4.8.3.158 WAVE_FORMAT_ZYXEL_ADPCM

```
readonly Guid WAVE_FORMAT_ZYXEL_ADPCM = new Guid((short)AudioEncoding.WAVE_FORMAT_ZYXEL_ADPCM & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_ZYXEL_ADPCM

4.8.3.159 WindowsMediaAudio

```
readonly Guid WindowsMediaAudio = new Guid((short)AudioEncoding.WindowsMediaAudio & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Windows Media Audio, WAVE_FORMAT_WMAUDIO2, Microsoft Corporation

4.8.3.160 WindowsMediaAudioLossless

```
readonly Guid WindowsMediaAudioLossless = new Guid((short)AudioEncoding.WindowsMediaAudioLossless & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Windows Media Audio Lossless, WAVE_FORMAT_WMAUDIO_LOSSLESS

4.8.3.161 WindowsMediaAudioProfessional

```
readonly Guid WindowsMediaAudioProfessional = new Guid((short)AudioEncoding.WindowsMediaAudioProfessional & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Windows Media Audio Professional WAVE_FORMAT_WMAUDIO3, Microsoft Corporation

4.8.3.162 WindowsMediaAudioSpdif

```
readonly Guid WindowsMediaAudioSpdif = new Guid((short)AudioEncoding.WindowsMediaAudioSpdif & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

Windows Media Audio Professional over SPDIF WAVE_FORMAT_WMASPDIF (0x0164)

4.8.3.163 WmaVoice9

```
readonly Guid WmaVoice9 = new Guid((short)AudioEncoding.WmaVoice9 & 0x0000FFFF, 0x0000, 0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_WMAVOICE9

4.8.3.164 YamahaAdpcm

```
readonly Guid YamahaAdpcm = new Guid((short)AudioEncoding.YamahaAdpcm & 0x0000FFFF, 0x0000,
0x0010, 0x80, 0x00, 0x00, 0xaa, 0x00, 0x38, 0x9b, 0x71) [static]
```

WAVE_FORMAT_YAMAHA_ADPCM Yamaha Corporation of America

4.9 AudioSyncBuffer< T > Class Template Reference

Inherits [IAudioOut< T >](#).

Public Member Functions

- **AudioSyncBuffer** (int playDelayMs, [ILogger](#) logger, string logPrefix, bool debugInfo)
- void **Start** (int sampleRate, int channels, int frameSamples)
- void **Service** ()
- void **Read** (T[] outBuf, int outChannels, int outSampleRate)
- void **Push** (T[] frame)
- void **Flush** ()
- void **Stop** ()

Static Public Attributes

- const int **FRAME_POOL_CAPACITY** = 50

Properties

- int **Lag** [get]
- bool **IsPlaying** [get]

4.10 AudioUtil Class Reference

Collection of Audio Utility functions and classes.

Classes

- interface [ILevelMeter](#)
Audio Level Metering interface.
- interface [IVoiceDetector](#)
Voice Activity Detector interface.
- class [LevelMeter](#)
Audio Level Meter.
- class [LevelMeterDummy](#)
Dummy Audio Level Meter that doesn't actually do anything.
- class [LevelMeterFloat](#)
LevelMeter specialization for float audio.
- class [LevelMeterShort](#)
LevelMeter specialization for short audio.
- class [Resampler](#)
Sample-rate conversion Audio Processor.
- class [ToneAudioPusher](#)
IAudioPusher that provides a constant tone signal.
- class [ToneAudioReader](#)
IAudioReader that provides a constant tone signal.
- class [VoiceDetector](#)
Simple voice activity detector triggered by signal level.
- class [VoiceDetectorCalibration](#)
Calibration Utility for Voice Detector
- class [VoiceDetectorDummy](#)
Dummy VoiceDetector that doesn't actually do anything.
- class [VoiceDetectorFloat](#)
VoiceDetector specialization for float audio.
- class [VoiceDetectorShort](#)
VoiceDetector specialization for float audio.
- class [VoiceLevelDetectCalibrate](#)
Utility Audio Processor Voice Detection Calibration.

Static Public Member Functions

- static void [Resample](#)< T > (T[] src, T[] dst, int dstCount, int channels)
Resample audio data so that the complete src buffer fits into dstCount samples in the dst buffer.
- static void **Resample**< T > (T[] src, int srcOffset, int srcCount, T[] dst, int dstOffset, int dstCount, int channels)
- static void **Resample**< T > (T[] src, int srcOffset, int srcCount, int srcChannels, T[] dst, int dstOffset, int dstCount, int dstChannels)
- static void [ResampleAndConvert](#) (short[] src, float[] dst, int dstCount, int channels)
Resample audio data so that the complete src buffer fits into dstCount samples in the dst buffer, and convert short to float samples along the way.
- static void [ResampleAndConvert](#) (float[] src, short[] dst, int dstCount, int channels)
Resample audio data so that the complete src buffer fits into dstCount samples in the dst buffer, and convert float to short samples along the way.
- static void [Convert](#) (float[] src, short[] dst, int dstCount)
Convert audio buffer from float to short samples.
- static void [Convert](#) (short[] src, float[] dst, int dstCount)
Convert audio buffer from short to float samples.
- static void [ForceToStereo](#)< T > (T[] src, T[] dst, int srcChannels)
Convert audio buffer with arbitrary number of channels to stereo.

4.10.1 Detailed Description

Collection of Audio Utility functions and classes.

4.10.2 Member Function Documentation

4.10.2.1 Convert() [1/2]

```
static void Convert (
    float[] src,
    short[] dst,
    int dstCount ) [static]
```

Convert audio buffer from float to short samples.

Parameters

<i>src</i>	Source buffer.
<i>dst</i>	Destination buffer.
<i>dstCount</i>	Size of destination buffer (in total samples), source buffer must be of same length or longer.

4.10.2.2 Convert() [2/2]

```
static void Convert (
    short[] src,
    float[] dst,
    int dstCount ) [static]
```

Convert audio buffer from short to float samples.

Parameters

<i>src</i>	Source buffer.
<i>dst</i>	Destination buffer.
<i>dstCount</i>	Size of destination buffer (in total samples), source buffer must be of same length or longer.

4.10.2.3 ForceToStereo< T >()

```
static void ForceToStereo< T > (
    T[] src,
```

```
T[] dst,
int srcChannels ) [static]
```

Convert audio buffer with arbitrary number of channels to stereo.

For mono sources (`srcChannels==1`), the signal will be copied to both Left and Right stereo channels. For all others, the first two available channels will be used, any other channels will be discarded.

Parameters

<i>src</i>	Source buffer.
<i>dst</i>	Destination buffer.
<i>srcChannels</i>	Number of (interleaved) channels in src.

4.10.2.4 Resample< T >()

```
static void Resample< T > (
    T[] src,
    T[] dst,
    int dstCount,
    int channels ) [static]
```

Resample audio data so that the complete src buffer fits into dstCount samples in the dst buffer.

This implements a primitive nearest-neighbor resampling algorithm for an arbitrary number of channels.

Parameters

<i>src</i>	Source buffer.
<i>dst</i>	Destination buffer.
<i>dstCount</i>	Target size of destination buffer (in samples per channel).
<i>channels</i>	Number of channels in the signal (1=mono, 2=stereo). Must be > 0.

4.10.2.5 ResampleAndConvert() [1/2]

```
static void ResampleAndConvert (
    float[] src,
    short[] dst,
    int dstCount,
    int channels ) [static]
```

Resample audio data so that the complete src buffer fits into dstCount samples in the dst buffer, and convert float to short samples along the way.

This implements a primitive nearest-neighbor resampling algorithm for an arbitrary number of channels.

Parameters

<i>src</i>	Source buffer.
<i>dst</i>	Destination buffer.
<i>dstCount</i>	Target size of destination buffer (in samples per channel).
<i>channels</i>	Number of channels in the signal (1=mono, 2=stereo). Must be > 0.

4.10.2.6 `ResampleAndConvert()` [2/2]

```
static void ResampleAndConvert (
    short[] src,
    float[] dst,
    int dstCount,
    int channels ) [static]
```

Resample audio data so that the complete src buffer fits into dstCount samples in the dst buffer, and convert short to float samples along the way.

This implements a primitive nearest-neighbor resampling algorithm for an arbitrary number of channels.

Parameters

<i>src</i>	Source buffer.
<i>dst</i>	Destination buffer.
<i>dstCount</i>	Target size of destination buffer (in samples per channel).
<i>channels</i>	Number of channels in the signal (1=mono, 2=stereo). Must be > 0.

4.11 `BufferReaderPushAdapter< T >` Class Template Reference

Simple [BufferReaderPushAdapterBase](#) implementation using a single buffer, using synchronous `LocalVoice.PushData`

Inherits [BufferReaderPushAdapterBase< T >](#).

Public Member Functions

- [BufferReaderPushAdapter](#) ([LocalVoice](#) localVoice, [IDataReader< T >](#) reader)
Create a new [BufferReaderPushAdapter](#) instance
- override void [Service](#) ([LocalVoice](#) localVoice)
Do the actual data read/push.

Protected Attributes

- `T[]` **buffer**

4.11.1 Detailed Description

Simple [BufferedReaderPushAdapterBase](#) implementation using a single buffer, using synchronous [LocalVoice.PushData](#)

4.11.2 Constructor & Destructor Documentation

4.11.2.1 [BufferedReaderPushAdapter\(\)](#)

```
BufferedReaderPushAdapter (
    LocalVoice localVoice,
    IDataReader< T > reader )
```

Create a new [BufferedReaderPushAdapter](#) instance

Parameters

<i>localVoice</i>	LocalVoice instance to push data to.
<i>reader</i>	IDataReader to read from.

4.11.3 Member Function Documentation

4.11.3.1 [Service\(\)](#)

```
override void Service (
    LocalVoice localVoice ) [virtual]
```

Do the actual data read/push.

Parameters

<i>localVoice</i>	LocalVoice instance to push data to.
-------------------	--

Implements [BufferedReaderPushAdapterBase< T >](#).

4.12 [BufferedReaderPushAdapterAsyncPool< T >](#) Class Template Reference

[BufferedReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#).

Inherits [BufferedReaderPushAdapterBase< T >](#).

Public Member Functions

- `BufferedReaderPushAdapterAsyncPool` (`LocalVoice` localVoice, `IDataReader`< T > reader)
Create a new `BufferedReaderPushAdapter` instance
- override void `Service` (`LocalVoice` localVoice)
Do the actual data read/push.

Additional Inherited Members

4.12.1 Detailed Description

`BufferedReaderPushAdapter` implementation using asynchronous `LocalVoice.PushDataAsync`.

Acquires a buffer from pool before each Read, releases buffer after last Read (brings Acquire/Release overhead).

Expects localVoice to be a `LocalVoiceFramed<T>` of same T.

4.12.2 Constructor & Destructor Documentation

4.12.2.1 `BufferedReaderPushAdapterAsyncPool()`

```
BufferedReaderPushAdapterAsyncPool (
    LocalVoice localVoice,
    IDataReader< T > reader )
```

Create a new `BufferedReaderPushAdapter` instance

Parameters

<code>localVoice</code>	<code>LocalVoice</code> instance to push data to.
<code>reader</code>	<code>DataReader</code> to read from.

4.12.3 Member Function Documentation

4.12.3.1 `Service()`

```
override void Service (
    LocalVoice localVoice ) [virtual]
```

Do the actual data read/push.

Parameters

<i>localVoice</i>	LocalVoice instance to push data to. Must be a LocalVoiceFramed<T> of same T.
-------------------	---

Implements [BufferReaderPushAdapterBase< T >](#).

4.13 BufferReaderPushAdapterAsyncPoolCopy< T > Class Template Reference

[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#) and data copy.

Inherits [BufferReaderPushAdapterBase< T >](#).

Public Member Functions

- [BufferReaderPushAdapterAsyncPoolCopy](#) ([LocalVoice](#) localVoice, [IDataReader< T >](#) reader)
Create a new [BufferReaderPushAdapter](#) instance
- override void [Service](#) ([LocalVoice](#) localVoice)
Do the actual data read/push.

Protected Attributes

- `T[]` `buffer`

4.13.1 Detailed Description

[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#) and data copy.

Reads data to preallocated buffer, copies it to buffer from pool before pushing. Compared with [, this avoids one pool Acquire/Release](#)

4.13.2 Constructor & Destructor Documentation

4.13.2.1 BufferReaderPushAdapterAsyncPoolCopy()

```
BufferReaderPushAdapterAsyncPoolCopy (
    LocalVoice localVoice,
    IDataReader< T > reader )
```

Create a new [BufferReaderPushAdapter](#) instance

Parameters

<i>localVoice</i>	LocalVoice instance to push data to.
<i>reader</i>	DataReader to read from.

4.13.3 Member Function Documentation

4.13.3.1 Service()

```
override void Service (
    LocalVoice localVoice ) [virtual]
```

Do the actual data read/push.

Parameters

<i>localVoice</i>	LocalVoice instance to push data to. Must be a LocalVoiceFramed<T> of same T.
-------------------	---

Implements [BufferReaderPushAdapterBase< T >](#).

4.14 BufferReaderPushAdapterAsyncPoolFloatToShort Class Reference

[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#), converting float samples to short.

Inherits [BufferReaderPushAdapterBase< float >](#).

Public Member Functions

- [BufferReaderPushAdapterAsyncPoolFloatToShort](#) ([LocalVoice](#) localVoice, [IDataReader](#)< float > reader)
Create a new [BufferReaderPushAdapter](#) instance
- override void [Service](#) ([LocalVoice](#) localVoice)
Do the actual data read/push.

Additional Inherited Members

4.14.1 Detailed Description

[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#), converting float samples to short.

This adapter works exactly like [BufferReaderPushAdapterAsyncPool](#), but it converts float samples to short. Acquires a buffer from pool before each Read, releases buffer after last Read.

Expects localVoice to be a [LocalVoiceFramed<T>](#) of same T.

4.14.2 Constructor & Destructor Documentation

4.14.2.1 `BufferReaderPushAdapterAsyncPoolFloatToShort()`

```
BufferReaderPushAdapterAsyncPoolFloatToShort (
    LocalVoice localVoice,
    IDataReader< float > reader )
```

Create a new [BufferReaderPushAdapter](#) instance

Parameters

<i>localVoice</i>	LocalVoice instance to push data to.
<i>reader</i>	IDataReader to read from.

4.14.3 Member Function Documentation

4.14.3.1 `Service()`

```
override void Service (
    LocalVoice localVoice ) [virtual]
```

Do the actual data read/push.

Parameters

<i>localVoice</i>	LocalVoice instance to push data to. Must be a LocalVoiceFramed<T> of same T.
-------------------	---

Implements [BufferReaderPushAdapterBase< float >](#).

4.15 `BufferReaderPushAdapterAsyncPoolShortToFloat` Class Reference

[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#), converting short samples to float.

Inherits [BufferReaderPushAdapterBase< short >](#).

Public Member Functions

- [BufferReaderPushAdapterAsyncPoolShortToFloat](#) ([LocalVoice](#) localVoice, [IDataReader](#)< short > reader)
Create a new [BufferReaderPushAdapter](#) instance
- override void [Service](#) ([LocalVoice](#) localVoice)
Do the actual data read/push.

Additional Inherited Members

4.15.1 Detailed Description

[BufferReaderPushAdapter](#) implementation using asynchronous [LocalVoice.PushDataAsync](#), converting short samples to float.

This adapter works exactly like [BufferReaderPushAdapterAsyncPool](#), but it converts short samples to float. Acquires a buffer from pool before each Read, releases buffer after last Read.

Expects localVoice to be a [LocalVoiceFramed<T>](#) of same T.

4.15.2 Constructor & Destructor Documentation

4.15.2.1 BufferReaderPushAdapterAsyncPoolShortToFloat()

```
BufferReaderPushAdapterAsyncPoolShortToFloat (
    LocalVoice localVoice,
    IDataReader< short > reader )
```

Create a new [BufferReaderPushAdapter](#) instance

Parameters

<i>localVoice</i>	LocalVoice instance to push data to.
<i>reader</i>	DataReader to read from.

4.15.3 Member Function Documentation

4.15.3.1 Service()

```
override void Service (
    LocalVoice localVoice ) [virtual]
```

Do the actual data read/push.

Parameters

<i>localVoice</i>	LocalVoice instance to push data to. Must be a LocalVoiceFramed<T> of same T.
-------------------	---

Implements [BufferReaderPushAdapterBase< short >](#).

4.16 **BufferedReaderPushAdapterBase< T > Class Template Reference**

Adapter base class to move data by reading from [IDataReader.Read](#) and pushing to [LocalVoice](#).

Inherits [IServiceable](#).

Inherited by [BufferedReaderPushAdapter< T >](#), [BufferedReaderPushAdapterAsyncPool< T >](#), and [BufferedReaderPushAdapterAsyncPool](#).

Public Member Functions

- abstract void [Service](#) ([LocalVoice](#) localVoice)
Do the actual data read/push.
- [BufferedReaderPushAdapterBase](#) ([IDataReader](#)< T > reader)
Create a new [BufferedReaderPushAdapterBase](#) instance
- void [Dispose](#) ()
Release resources associated with this instance.

Protected Attributes

- [IDataReader](#)< T > **reader**

4.16.1 Detailed Description

Adapter base class to move data by reading from [IDataReader.Read](#) and pushing to [LocalVoice](#).

Use this with a [LocalVoice](#) of same T type.

4.16.2 Constructor & Destructor Documentation

4.16.2.1 [BufferedReaderPushAdapterBase\(\)](#)

```
BufferedReaderPushAdapterBase (  
    IDataReader< T > reader )
```

Create a new [BufferedReaderPushAdapterBase](#) instance

Parameters

<i>reader</i>	DataReader to read from.
---------------	--------------------------

4.16.3 Member Function Documentation

4.16.3.1 Dispose()

```
void Dispose ( )
```

Release resources associated with this instance.

4.16.3.2 Service()

```
abstract void Service (
    LocalVoice localVoice ) [pure virtual]
```

Do the actual data read/push.

Parameters

<i>localVoice</i>	LocalVoice instance to push data to.
-------------------	--

Implements [IServiceable](#).

Implemented in [BufferedReaderPushAdapterAsyncPoolShortToFloat](#), [BufferedReaderPushAdapterAsyncPoolFloatToShort](#), [BufferedReaderPushAdapterAsyncPoolCopy< T >](#), [BufferedReaderPushAdapterAsyncPool< T >](#), and [BufferedReaderPushAdapter< T >](#).

4.17 ConnectAndJoin Class Reference

Inherits [MonoBehaviour](#), [IConnectionCallbacks](#), and [IMatchmakingCallbacks](#).

Public Member Functions

- void **ConnectNow** ()
- void **OnCreatedRoom** ()
- void **OnCreateRoomFailed** (short returnCode, string message)
- void **OnFriendListUpdate** (List< FriendInfo > friendList)
- void **OnJoinedRoom** ()
- void **OnJoinRandomFailed** (short returnCode, string message)
- void **OnJoinRoomFailed** (short returnCode, string message)
- void **OnLeftRoom** ()
- void **OnConnected** ()
- void **OnConnectedToMaster** ()
- void **OnDisconnected** (DisconnectCause cause)
- void **OnRegionListReceived** (RegionHandler regionHandler)
- void **OnCustomAuthenticationResponse** (Dictionary< string, object > data)
- void **OnCustomAuthenticationFailed** (string debugMessage)

Public Attributes

- bool **RandomRoom** = true
- string **RoomName**

Properties

- bool **IsConnected** [get]

4.18 OpusCodec.Decoder< T > Class Template Reference

Inherits [IDecoder](#).

Public Member Functions

- **Decoder** (Action< [FrameOut](#)< T >> output, [ILogger](#) logger)
- void **Open** ([VoiceInfo](#) i)
Open (initialize) the decoder.
- void **Dispose** ()
- void **Input** (byte[] buf, FrameFlags flags)
Consumes the given encoded data.

Protected Attributes

- [OpusDecoder](#)< T > **decoder**

Properties

- string **Error** [get]

4.18.1 Member Function Documentation

4.18.1.1 Input()

```
void Input (
    byte[] buf,
    FrameFlags flags )
```

Consumes the given encoded data.

Implements [IDecoder](#).

4.18.1.2 Open()

```
void Open (
    VoiceInfo info )
```

Open (initialize) the decoder.

Parameters

<i>info</i>	Properties of the data stream to decode.
-------------	--

Implements [IDecoder](#).

4.19 RawCodec.Decoder< T > Class Template Reference

Inherits [IDecoder](#).

Public Member Functions

- **Decoder** (Action< [FrameOut](#)< T >> output)
- void [Open](#) ([VoiceInfo](#) info)
Open (initialize) the decoder.
- void [Input](#) (byte[] byteBuf, FrameFlags flags)
Consumes the given encoded data.
- void **Dispose** ()

Properties

- string **Error** [get]

4.19.1 Member Function Documentation

4.19.1.1 Input()

```
void Input (
    byte[] buf,
    FrameFlags flags )
```

Consumes the given encoded data.

Implements [IDecoder](#).

4.19.1.2 Open()

```
void Open (
    VoiceInfo info )
```

Open (initialize) the decoder.

Parameters

<i>info</i>	Properties of the data stream to decode.
-------------	--

Implements [IDecoder](#).

4.20 OpusCodec.DecoderFactory Class Reference

Static Public Member Functions

- static [IEncoder](#) **Create**< T > ([VoiceInfo](#) i, [ILogger](#) logger)

4.21 OpusCodec.Encoder< T > Class Template Reference

Inherits [IEncoderDirect< T\[\]>](#).

Public Member Functions

- void **Input** (T[] buf)
- void **EndOfStream** ()
- [ArraySegment< byte >](#) **DequeueOutput** (out [FrameFlags](#) flags)
- [I GetPlatformAPI](#)< I > ()
- void **Dispose** ()

Protected Member Functions

- **Encoder** ([VoiceInfo](#) i, [ILogger](#) logger)
- abstract [ArraySegment< byte >](#) **encodeTyped** (T[] buf)

Protected Attributes

- [OpusEncoder](#) **encoder**
- bool **disposed**

Properties

- string **Error** [get]
- [Action< ArraySegment< byte >, FrameFlags >](#) **Output** [get, set]

4.22 RawCodec.Encoder< T > Class Template Reference

Inherits [IEncoderDirect< T\[\]>](#).

Public Member Functions

- `ArraySegment< byte > DequeueOutput` (out FrameFlags flags)
- `void EndOfStream` ()
- `I GetPlatformAPI< I > ()`
- `void Dispose` ()
- `void Input` (T[] buf)

Properties

- `string Error` [get]
- `Action< ArraySegment< byte >, FrameFlags > Output` [get, set]

4.23 OpusCodec.EncoderFloat Class Reference

Inherits [OpusCodec.Encoder< float >](#).

Protected Member Functions

- `override ArraySegment< byte > encodeTyped` (float[] buf)

Additional Inherited Members

4.24 OpusCodec.EncoderShort Class Reference

Inherits [OpusCodec.Encoder< short >](#).

Protected Member Functions

- `override ArraySegment< byte > encodeTyped` (short[] buf)

Additional Inherited Members

4.25 Extensions Class Reference

Provides a few basic extensions.

4.25.1 Detailed Description

Provides a few basic extensions.

4.26 OpusCodec.Factory Class Reference

Static Public Member Functions

- static [IEncoder](#) **CreateEncoder**< [B](#) > ([VoiceInfo](#) i, [ILogger](#) logger)

4.27 FactoryPrimitiveArrayPool< T > Class Template Reference

PrimitiveArrayPool<T> as wrapped in object factory interface.

Inherits [ObjectFactory](#)< [T\[\]](#), [int](#) >.

Public Member Functions

- **FactoryPrimitiveArrayPool** (int capacity, string name)
- **FactoryPrimitiveArrayPool** (int capacity, string name, int info)
- [T\[\]](#) **New** ()
- [T\[\]](#) **New** (int size)
- void **Free** ([T\[\]](#) obj)
- void **Free** ([T\[\]](#) obj, int info)
- void **Dispose** ()

Properties

- int **Info** [get]

4.27.1 Detailed Description

PrimitiveArrayPool<T> as wrapped in object factory interface.

Template Parameters

T	Array element type.
-------------------	---------------------

4.28 FactoryReusableArray< T > Class Template Reference

Array factory returnig the same array instance as long as it requested with the same array length. If length changes, new array instance created.

Inherits [ObjectFactory](#)< [T\[\]](#), [int](#) >.

Public Member Functions

- **FactoryReusableArray** (int size)

- **T[] New** ()
- **T[] New** (int size)
- void **Free** (T[] obj)
- void **Free** (T[] obj, int info)
- void **Dispose** ()

Properties

- int **Info** [get]

4.28.1 Detailed Description

Array factory returnig the same array instance as long as it requested with the same array length. If length changes, new array instance created.

Template Parameters

<i>T</i>	Array element type.
----------	---------------------

4.29 FrameOut< T > Class Template Reference

Public Member Functions

- **FrameOut** (T[] buf, bool endOfStream)
- **FrameOut**< T > **Set** (T[] buf, bool endOfStream)

Properties

- **T[] Buf** [get]
- bool **EndOfStream** [get]

4.30 Framer< T > Class Template Reference

Utility class to re-frame audio packets.

Public Member Functions

- **Framer** (int frameSize)
*Create new **Framer** instance.*
- int **Count** (int bufLen)
Get the number of frames available after adding bufLen samples.
- IEnumerable< T[] > **Frame** (T[] buf)
Append arbitrary-sized buffer and return available full frames.

4.30.1 Detailed Description

Utility class to re-frame audio packets.

4.30.2 Constructor & Destructor Documentation

4.30.2.1 Framer()

```
Framer (
    int frameSize )
```

Create new [Framer](#) instance.

4.30.3 Member Function Documentation

4.30.3.1 Count()

```
int Count (
    int bufLen )
```

Get the number of frames available after adding *bufLen* samples.

Parameters

<i>bufLen</i>	Number of samples that would be added.
---------------	--

Returns

Number of full frames available when adding *bufLen* samples.

4.30.3.2 Frame()

```
IEnumerable<T[]> Frame (
    T[] buf )
```

Append arbitrary-sized buffer and return available full frames.

Parameters

<i>buf</i>	Array of samples to add.
------------	--------------------------

Returns

Enumerator of full frames (might be none).

4.31 IAudioDesc Interface Reference

Audio Source interface.

Inherits IDisposable.

Inherited by [AudioDesc](#), [IAudioPusher< T >](#), and [IAudioReader< T >](#).

Properties

- int [SamplingRate](#) [get]
Sampling rate of the audio signal (in Hz).
- int [Channels](#) [get]
Number of channels in the audio signal.
- string [Error](#) [get]
If not null, audio object is in invalid state.

4.31.1 Detailed Description

Audio Source interface.

4.31.2 Property Documentation

4.31.2.1 Channels

```
int Channels [get]
```

Number of channels in the audio signal.

4.31.2.2 Error

```
string Error [get]
```

If not null, audio object is in invalid state.

4.31.2.3 SamplingRate

```
int SamplingRate [get]
```

Sampling rate of the audio signal (in Hz).

4.32 IAudioOut< T > Interface Template Reference

Inherited by [AudioSyncBuffer< T >](#).

Public Member Functions

- void **Start** (int frequency, int channels, int frameSamplesPerChannel)
- void **Flush** ()
- void **Stop** ()
- void **Push** (T[] frame)
- void **Service** ()

Properties

- bool **IsPlaying** [get]
- int **Lag** [get]

4.33 IAudioPusher< T > Interface Template Reference

Audio Pusher interface.

Inherits [IAudioDesc](#).

Inherited by [AudioUtil.ToneAudioPusher< T >](#).

Public Member Functions

- void [SetCallback](#) (Action< T[]> callback, [ObjectFactory](#)< T[], int > bufferFactory)
Set the callback function used for pushing data.

Additional Inherited Members

4.33.1 Detailed Description

Audio Pusher interface.

Opposed to an [IAudioReader](#) (which will deliver audio data when it is "pulled"), an [IAudioPusher](#) will push its audio data whenever it is ready,

4.33.2 Member Function Documentation

4.33.2.1 SetCallback()

```
void SetCallback (
    Action< T[]> callback,
    ObjectFactory< T[], int > bufferFactory )
```

Set the callback function used for pushing data.

Parameters

<i>callback</i>	Callback function to use.
<i>localVoice</i>	Outgoing audio stream, for context.

Implemented in [AudioUtil.ToneAudioPusher< T >](#).

4.34 IAudioReader< T > Interface Template Reference

Audio Reader interface.

Inherits [IDataReader< T >](#), and [IAudioDesc](#).

Inherited by [AudioUtil.ToneAudioReader< T >](#).

Additional Inherited Members

4.34.1 Detailed Description

Audio Reader interface.

Opposed to an [IAudioPusher](#) (which will push its audio data whenever it is ready), an [IAudioReader](#) will deliver audio data when it is "pulled" (it's Read function is called).

4.35 IAudioSource Interface Reference

Defines the base for all audio streams.

Inherits [IDisposable](#).

Inherited by [IReadableAudioSource< in in T >](#).

Properties

- bool [CanSeek](#) [get]
Gets a value indicating whether the [IAudioSource](#) supports seeking.
- [WaveFormat](#) [WaveFormat](#) [get]
Gets the [WaveFormat](#) of the waveform-audio data.
- long [Position](#) [get, set]
Gets or sets the current position. The unit of this property depends on the implementation of this interface. Some implementations may not support this property.
- long [Length](#) [get]
Gets the length of the waveform-audio data. The unit of this property depends on the implementation of this interface. Some implementations may not support this property.

4.35.1 Detailed Description

Defines the base for all audio streams.

4.35.2 Property Documentation

4.35.2.1 CanSeek

```
bool CanSeek [get]
```

Gets a value indicating whether the [IAudioSource](#) supports seeking.

4.35.2.2 Length

```
long Length [get]
```

Gets the length of the waveform-audio data. The unit of this property depends on the implementation of this interface. Some implementations may not support this property.

4.35.2.3 Position

```
long Position [get], [set]
```

Gets or sets the current position. The unit of this property depends on the implementation of this interface. Some implementations may not support this property.

4.35.2.4 WaveFormat

```
WaveFormat WaveFormat [get]
```

Gets the [WaveFormat](#) of the waveform-audio data.

4.36 `IDataReader< T >` Interface Template Reference

Interface for pulling data, in case this is more appropriate than pushing it.

Inherits `IDisposable`.

Inherited by [IAudioReader< T >](#).

Public Member Functions

- bool [Read](#) (T[] buffer)

Fill full given frame buffer with source uncompressed data or return false if not enough such data.

4.36.1 Detailed Description

Interface for pulling data, in case this is more appropriate than pushing it.

4.36.2 Member Function Documentation

4.36.2.1 Read()

```
bool Read (
    T[] buffer )
```

Fill full given frame buffer with source uncompressed data or return false if not enough such data.

Parameters

<i>buffer</i>	Buffer to fill.
---------------	-----------------

Returns

True if buffer was filled successfully, false otherwise.

Implemented in [AudioUtil.ToneAudioReader< T >](#).

4.37 IDecoder Interface Reference

Generic decoder interface.

Inherits [IDisposable](#).

Inherited by [IDecoderDirect< B >](#), [OpusCodec.Decoder< T >](#), and [RawCodec.Decoder< T >](#).

Public Member Functions

- void [Open](#) ([VoiceInfo](#) info)
Open (initialize) the decoder.
- void [Input](#) (byte[] buf, [FrameFlags](#) flags)
Consumes the given encoded data.

Properties

- string [Error](#) [get]
If not null, the object is in invalid state.

4.37.1 Detailed Description

Generic decoder interface.

4.37.2 Member Function Documentation

4.37.2.1 Input()

```
void Input (
    byte[] buf,
    FrameFlags flags )
```

Consumes the given encoded data.

Implemented in [RawCodec.Decoder< T >](#), and [OpusCodec.Decoder< T >](#).

4.37.2.2 Open()

```
void Open (
    VoiceInfo info )
```

Open (initialize) the decoder.

Parameters

<i>info</i>	Properties of the data stream to decode.
-------------	--

Implemented in [RawCodec.Decoder< T >](#), and [OpusCodec.Decoder< T >](#).

4.37.3 Property Documentation

4.37.3.1 Error

```
string Error [get]
```

If not null, the object is in invalid state.

4.38 IDecoderDirect< B > Interface Template Reference

Interface for an decoder which outputs data via explicit call.

Inherits [IDecoder](#).

Properties

- Action< B > **Output** [get, set]

Additional Inherited Members

4.38.1 Detailed Description

Interface for an decoder which outputs data via explicit call.

4.39 IDecoderQueuedOutputImageNative Interface Reference

Inherits [IDecoderDirect< ImageOutputBuf >](#).

Properties

- ImageFormat **OutputImageFormat** [get, set]
- Func< int, int, IntPtr > **OutputImageBufferGetter** [get, set]

4.40 IEncoder Interface Reference

Generic encoder interface.

Inherits [IDisposable](#).

Inherited by [IEncoderDirect< B >](#).

Public Member Functions

- ArraySegment< byte > [DequeueOutput](#) (out FrameFlags flags)
Returns next encoded data frame (if such output supported).
- void [EndOfStream](#) ()
Forces an encoder to flush and produce frame with EndOfStream flag (in output queue).
- I **GetPlatformAPI**< I > ()

Properties

- string [Error](#) [get]
If not null, the object is in invalid state.
- Action< ArraySegment< byte >, FrameFlags > [Output](#) [set]
Set callback encoder calls on each encoded data frame (if such output supported).

4.40.1 Detailed Description

Generic encoder interface.

Depending on implementation, encoder should either call Output on eaach data frame or return next data frame in [DequeueOutput\(\)](#) call.

4.40.2 Member Function Documentation

4.40.2.1 DequeueOutput()

```
ArraySegment<byte> DequeueOutput (
    out FrameFlags flags )
```

Returns next encoded data frame (if such output supported).

4.40.2.2 EndOfStream()

```
void EndOfStream ( )
```

Forces an encoder to flush and produce frame with EndOfStream flag (in output queue).

4.40.3 Property Documentation

4.40.3.1 Error

```
string Error [get]
```

If not null, the object is in invalid state.

4.40.3.2 Output

`Action<ArraySegment<byte>, FrameFlags> Output [set]`

Set callback encoder calls on each encoded data frame (if such output supported).

4.41 IEncoderDirect< B > Interface Template Reference

Interface for an encoder which consumes input data via explicit call.

Inherits [IEncoder](#).

Public Member Functions

- void [Input](#) (B buf)
Consumes the given raw data.

Additional Inherited Members

4.41.1 Detailed Description

Interface for an encoder which consumes input data via explicit call.

4.41.2 Member Function Documentation

4.41.2.1 Input()

```
void Input (
    B buf )
```

Consumes the given raw data.

Parameters

<i>buf</i>	Array containing raw data (e.g. audio samples).
------------	---

4.42 AudioUtil.ILevelMeter Interface Reference

Audio Level Metering interface.

Inherited by [AudioUtil.LevelMeter< T >](#), and [AudioUtil.LevelMeterDummy](#).

Public Member Functions

- void [ResetAccumAvgPeakAmp](#) ()
Reset [AccumAvgPeakAmp](#).

Properties

- float [CurrentAvgAmp](#) [get]
Average amplitude value over last half second.
- float [CurrentPeakAmp](#) [get]
Maximum amplitude value over last half second sec.
- float [AccumAvgPeakAmp](#) [get]
Average of CurrentPeakAmps since last reset.

4.42.1 Detailed Description

Audio Level Metering interface.

4.42.2 Member Function Documentation

4.42.2.1 ResetAccumAvgPeakAmp()

```
void ResetAccumAvgPeakAmp ( )
```

Reset [AccumAvgPeakAmp](#).

Implemented in [AudioUtil.LevelMeter< T >](#), and [AudioUtil.LevelMeterDummy](#).

4.42.3 Property Documentation

4.42.3.1 AccumAvgPeakAmp

```
float AccumAvgPeakAmp [get]
```

Average of CurrentPeakAmps since last reset.

4.42.3.2 CurrentAvgAmp

`float CurrentAvgAmp [get]`

Average amplitude value over last half second.

4.42.3.3 CurrentPeakAmp

`float CurrentPeakAmp [get]`

Maximum amplitude value over last half second sec.

4.43 ILocalVoiceAudio Interface Reference

Interface for an outgoing audio stream.

Inherited by [LocalVoiceAudio< T >](#), and [LocalVoiceAudioDummy](#).

Public Member Functions

- void [VoiceDetectorCalibrate](#) (int durationMs, Action< float > onCalibrated=null)
Trigger voice detector calibration process.

Properties

- [AudioUtil.IVoiceDetector VoiceDetector](#) [get]
The VoiceDetector in use.
- [AudioUtil.ILevelMeter LevelMeter](#) [get]
The LevelMeter utility in use.
- bool [VoiceDetectorCalibrating](#) [get]
If true, voice detector calibration is in progress.

4.43.1 Detailed Description

Interface for an outgoing audio stream.

A [LocalVoice](#) always brings a LevelMeter and a VoiceDetector, which you can access using this interface.

4.43.2 Member Function Documentation

4.43.2.1 VoiceDetectorCalibrate()

```
void VoiceDetectorCalibrate (
    int durationMs,
    Action< float > onCalibrated = null )
```

Trigger voice detector calibration process.

While calibrating, keep silence. [Voice](#) detector sets threshold based on measured background noise level.

Parameters

<i>durationMs</i>	Duration of calibration (in milliseconds).
<i>onCalibrated</i>	Called when calibration is complete. Parameter is new threshold value.

Implemented in [LocalVoiceAudioDummy](#), and [LocalVoiceAudio< T >](#).

4.43.3 Property Documentation

4.43.3.1 LevelMeter

[AudioUtil.ILoggable](#) LevelMeter [get]

The LevelMeter utility in use.

4.43.3.2 VoiceDetector

[AudioUtil.ILoggable](#) VoiceDetector [get]

The VoiceDetector in use.

Use it to enable or disable voice detector and set its parameters.

4.43.3.3 VoiceDetectorCalibrating

bool VoiceDetectorCalibrating [get]

If true, voice detector calibration is in progress.

4.44 ILoggable Interface Reference

Inherited by [ILoggableDependent](#), and [VoiceConnection](#).

Properties

- DebugLevel **LogLevel** [get, set]
- [VoiceLogger](#) **Logger** [get]

4.45 ILoggableDependent Interface Reference

Inherits [ILoggable](#).

Inherited by [VoiceComponent](#).

Properties

- bool **IgnoreGlobalLogLevel** [get, set]

4.46 ILogger Interface Reference

Inherited by [IVoiceTransport](#), [Logger](#), and [VoiceLogger](#).

Public Member Functions

- void **LogError** (string fmt, params object[] args)
- void **LogWarning** (string fmt, params object[] args)
- void **LogInfo** (string fmt, params object[] args)
- void **LogDebug** (string fmt, params object[] args)

4.47 ImageBufferInfo Class Reference

Public Member Functions

- **ImageBufferInfo** (int width, int height, int[] stride, ImageFormat format)

Properties

- int **Width** [get]
- int **Height** [get]
- int[] **Stride** [get]
- ImageFormat **Format** [get]
- Rotation **Rotation** [get, set]
- Flip **Flip** [get, set]

4.48 ImageBufferNative Class Reference

Inherited by [ImageBufferNativeAlloc](#), and [ImageBufferNativeGCHandleSinglePlane](#).

Public Member Functions

- **ImageBufferNative** ([ImageBufferInfo](#) info)
- virtual void **Release** ()
- virtual void **Dispose** ()

Properties

- [ImageBufferInfo](#) **Info** [get, protected set]
- [IntPtr\[\]](#) **Planes** [get, protected set]

4.49 ImageBufferNativeAlloc Class Reference

Inherits [ImageBufferNative](#), and [IDisposable](#).

Public Member Functions

- **ImageBufferNativeAlloc** ([ImageBufferNativePool](#)< [ImageBufferNativeAlloc](#) > pool, [ImageBufferInfo](#) info)
- override void **Release** ()
- override void **Dispose** ()

Additional Inherited Members

4.50 ImageBufferNativeGCHandleSinglePlane Class Reference

Inherits [ImageBufferNative](#), and [IDisposable](#).

Public Member Functions

- **ImageBufferNativeGCHandleSinglePlane** ([ImageBufferNativePool](#)< [ImageBufferNativeGCHandleSinglePlane](#) > pool, [ImageBufferInfo](#) info)
- void **PinPlane** (byte[] plane)
- override void **Release** ()
- override void **Dispose** ()

Additional Inherited Members

4.51 ImageBufferNativePool< T > Class Template Reference

Inherits [ObjectPool](#)< [T](#), [ImageBufferInfo](#) >.

Public Member Functions

- delegate [T](#) **Factory** ([ImageBufferNativePool](#)< [T](#) > pool, [ImageBufferInfo](#) info)
- **ImageBufferNativePool** (int capacity, [Factory](#) factory, string name)
- **ImageBufferNativePool** (int capacity, [Factory](#) factory, string name, [ImageBufferInfo](#) info)

Protected Member Functions

- override T **createObject** ([ImageBufferInfo](#) info)
- override void **destroyObject** (T obj)
- override bool **infosMatch** ([ImageBufferInfo](#) i0, [ImageBufferInfo](#) i1)

Additional Inherited Members

4.52 ImageInputBuf Struct Reference

Public Attributes

- [IntPtr](#)[] **Buf**
- int **Width**
- int **Height**
- [int](#)[] **Stride**
- [ImageFormat](#) **ImageFormat**
- [Rotation](#) **Rotation**
- [Flip](#) **Flip**

4.53 ImageOutputBuf Struct Reference

Public Attributes

- [IntPtr](#) **Buf**
- int **Width**
- int **Height**
- int **Stride**

4.54 IOSAudioForceToSpeaker Class Reference

Inherits [MonoBehaviour](#).

4.55 IProcessor< T > Interface Template Reference

Audio Processor interface.

Inherits [IDisposable](#).

Inherited by [AudioUtil.LevelMeter< T >](#), [AudioUtil.Resampler< T >](#), [AudioUtil.VoiceDetector< T >](#), [AudioUtil.VoiceDetectorCalibrate< T >](#) and [AudioUtil.VoiceLevelDetectCalibrate< T >](#).

Public Member Functions

- [T](#)[] **Process** ([T](#)[] buf)
Process a frame of audio data.

4.55.1 Detailed Description

Audio Processor interface.

4.55.2 Member Function Documentation

4.55.2.1 Process()

```
T [] Process (
    T[] buf )
```

Process a frame of audio data.

Parameters

<i>buf</i>	Buffer containing input audio data
------------	------------------------------------

Returns

Buffer containing output audio data or null if frame has been discarded (VAD)

Implemented in [AudioUtil.VoiceLevelDetectCalibrate< T >](#), [AudioUtil.VoiceDetector< T >](#), [AudioUtil.VoiceDetectorCalibration< T >](#), [AudioUtil.LevelMeter< T >](#), and [AudioUtil.Resampler< T >](#).

4.56 IReadableAudioSource< in in T > Interface Template Reference

Defines a generic base for all readable audio streams.

Inherits [IAudioSource](#).

Public Member Functions

- int [Read](#) (T[] buffer, int offset, int count)

Reads a sequence of elements from the IReadableAudioSource< T> and advances the position within the stream by the number of elements read.

Additional Inherited Members

4.56.1 Detailed Description

Defines a generic base for all readable audio streams.

Template Parameters

<i>T</i>	The type of the provided audio data.
----------	--------------------------------------

4.56.2 Member Function Documentation

4.56.2.1 Read()

```
int Read (
    T[] buffer,
    int offset,
    int count )
```

Reads a sequence of elements from the `IReadableAudioSource<T>` and advances the position within the stream by the number of elements read.

Parameters

<i>buffer</i>	An array of elements. When this method returns, the <i>buffer</i> contains the specified array of elements with the values between <i>offset</i> and (<i>offset</i> + <i>count</i> - 1) replaced by the elements read from the current source.
<i>offset</i>	The zero-based offset in the <i>buffer</i> at which to begin storing the data read from the current stream.
<i>count</i>	The maximum number of elements to read from the current source.

Returns

The total number of elements read into the buffer.

4.57 IResettable Interface Reference

Inherited by [UnityAndroidAudioInAEC](#).

Public Member Functions

- void **Reset** ()

4.58 IServiceable Interface Reference

Interface for classes that want their [Service\(\)](#) function to be called regularly in the context of a [LocalVoice](#).

Inherited by [BufferReaderPushAdapterBase< T >](#).

Public Member Functions

- void [Service](#) ([LocalVoice](#) localVoice)
Service function that should be called regularly.

4.58.1 Detailed Description

Interface for classes that want their [Service\(\)](#) function to be called regularly in the context of a [LocalVoice](#).

4.58.2 Member Function Documentation

4.58.2.1 Service()

```
void Service (  
    LocalVoice localVoice )
```

Service function that should be called regularly.

Implemented in [BufferReaderPushAdapterAsyncPoolCopy< T >](#), [BufferReaderPushAdapterAsyncPool< T >](#), [BufferReaderPushAdapter< T >](#), and [BufferReaderPushAdapterBase< T >](#).

4.59 AudioUtil.IVoiceDetector Interface Reference

[Voice](#) Activity Detector interface.

Inherited by [AudioUtil.VoiceDetector< T >](#), and [AudioUtil.VoiceDetectorDummy](#).

Properties

- bool [On](#) [get, set]
If true, voice detection enabled.
- float [Threshold](#) [get, set]
Voice detected as soon as signal level exceeds threshold.
- bool [Detected](#) [get]
If true, voice detected.
- DateTime [DetectedTime](#) [get]
Last time when switched to detected state.
- int [ActivityDelayMs](#) [get, set]
Keep detected state during this time after signal level dropped below threshold.

Events

- Action [OnDetected](#)
Called when switched to detected state.

4.59.1 Detailed Description

[Voice](#) Activity Detector interface.

4.59.2 Property Documentation

4.59.2.1 ActivityDelayMs

```
int ActivityDelayMs [get], [set]
```

Keep detected state during this time after signal level dropped below threshold.

4.59.2.2 Detected

```
bool Detected [get]
```

If true, voice detected.

4.59.2.3 DetectedTime

```
DateTime DetectedTime [get]
```

Last time when switched to detected state.

4.59.2.4 On

```
bool On [get], [set]
```

If true, voice detection enabled.

4.59.2.5 Threshold

```
float Threshold [get], [set]
```

[Voice](#) detected as soon as signal level exceeds threshold.

4.59.3 Event Documentation

4.59.3.1 OnDetected

Action OnDetected

Called when switched to detected state.

4.60 IVoiceTransport Interface Reference

Inherits [ILogger](#).

Inherited by [LoadBalancingTransport](#).

Public Member Functions

- bool **IsChannelJoined** (int channelId)
- void **SendVoicesInfo** (IEnumerable< [LocalVoice](#) > voices, int channelId, int targetPlayerId)
- void **SendVoiceRemove** ([LocalVoice](#) voice, int channelId, int targetPlayerId)
- void **SendFrame** (ArraySegment< byte > data, FrameFlags flags, byte evNumber, byte voiceld, int channelId, int targetPlayerId, bool reliable, [LocalVoice](#) localVoice)
- string **ChannelIdStr** (int channelId)
- string **PlayerIdStr** (int playerId)

4.61 IWaveSource Interface Reference

Defines the base for all audio streams which provide raw byte data.

Inherits [IReadableAudioSource< byte >](#).

Additional Inherited Members

4.61.1 Detailed Description

Defines the base for all audio streams which provide raw byte data.

Compared to the [ISampleSource](#), the [IWaveSource](#) provides raw bytes instead of samples. That means that the [IAudioSource.Position](#) and the [IAudioSource.Position](#) properties are expressed in bytes. Also the [IReadableAudioSource<T>.Read](#) method provides samples instead of raw bytes.

4.62 IWriteable Interface Reference

Provides the [Write](#) method.

Inherited by [WaveWriter](#).

Public Member Functions

- void [Write](#) (byte[] buffer, int offset, int count)
Used to write down raw byte data.

4.62.1 Detailed Description

Provides the [Write](#) method.

4.62.2 Member Function Documentation

4.62.2.1 Write()

```
void Write (  
    byte[] buffer,  
    int offset,  
    int count )
```

Used to write down raw byte data.

Parameters

<i>buffer</i>	Byte array which contains the data to write down.
<i>offset</i>	Zero-based offset in the <i>buffer</i> .
<i>count</i>	Number of bytes to write.

Implemented in [WaveWriter](#).

4.63 AudioUtil.LevelMeter< T > Class Template Reference

Audio Level Meter.

Inherits [IProcessor< T >](#), and [AudioUtil.ILevelMeter](#).

Public Member Functions

- void [ResetAccumAvgPeakAmp](#) ()
Reset AccumAvgPeakAmp.
- abstract T[] [Process](#) (T[] buf)
Process a frame of audio data.
- void **Dispose** ()

Protected Attributes

- float **ampSum**
- float **ampPeak**
- int **bufferSize**
- float[] **prevValues**
- int **prevValuesHead**
- float **accumAvgPeakAmpSum**
- int **accumAvgPeakAmpCount**
- float **currentPeakAmp**
- float **norm**

Properties

- float **CurrentAvgAmp** [get]
- float **CurrentPeakAmp** [get, protected set]
- float? **AccumAvgPeakAmp** [get]

4.63.1 Detailed Description

Audio Level Meter.

4.63.2 Member Function Documentation

4.63.2.1 Process()

```
abstract T [] Process (
    T[] buf ) [pure virtual]
```

Process a frame of audio data.

Parameters

<i>buf</i>	Buffer containing input audio data
------------	------------------------------------

Returns

Buffer containing output audio data or null if frame has been discarded (VAD)

Implements [IProcessor< T >](#).

4.63.2.2 ResetAccumAvgPeakAmp()

```
void ResetAccumAvgPeakAmp ( )
```

Reset AccumAvgPeakAmp.

Implements [AudioUtil.ILevelMeter](#).

4.64 AudioUtil.LevelMeterDummy Class Reference

Dummy Audio Level Meter that doesn't actually do anything.

Inherits [AudioUtil.ILevelMeter](#).

Public Member Functions

- void [ResetAccumAvgPeakAmp](#) ()
Reset AccumAvgPeakAmp.

Properties

- float **CurrentAvgAmp** [get]
- float **CurrentPeakAmp** [get]
- float **AccumAvgPeakAmp** [get]

4.64.1 Detailed Description

Dummy Audio Level Meter that doesn't actually do anything.

4.64.2 Member Function Documentation

4.64.2.1 ResetAccumAvgPeakAmp()

```
void ResetAccumAvgPeakAmp ( )
```

Reset AccumAvgPeakAmp.

Implements [AudioUtil.ILevelMeter](#).

4.65 AudioUtil.LevelMeterFloat Class Reference

[LevelMeter](#) specialization for float audio.

Inherits [AudioUtil.LevelMeter< float >](#).

Public Member Functions

- [LevelMeterFloat](#) (int samplingRate, int numChannels)
Create new [LevelMeterFloat](#) instance.
- override float[] **Process** (float[] buf)

Additional Inherited Members

4.65.1 Detailed Description

[LevelMeter](#) specialization for float audio.

4.65.2 Constructor & Destructor Documentation

4.65.2.1 LevelMeterFloat()

```
LevelMeterFloat (
    int samplingRate,
    int numChannels )
```

Create new [LevelMeterFloat](#) instance.

Parameters

<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>numChannels</i>	Number of channels in the audio signal.

4.66 AudioUtil.LevelMeterShort Class Reference

[LevelMeter](#) specialization for short audio.

Inherits [AudioUtil.LevelMeter< short >](#).

Public Member Functions

- [LevelMeterShort](#) (int samplingRate, int numChannels)
Create new [LevelMeterShort](#) instance.
- override short[] **Process** (short[] buf)

Additional Inherited Members

4.66.1 Detailed Description

[LevelMeter](#) specialization for short audio.

4.66.2 Constructor & Destructor Documentation

4.66.2.1 LevelMeterShort()

```
LevelMeterShort (
    int samplingRate,
    int numChannels )
```

Create new [LevelMeterShort](#) instance.

Parameters

<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>numChannels</i>	Number of channels in the audio signal.

4.67 LoadBalancingFrontend Class Reference

Inherits [LoadBalancingTransport](#).

Additional Inherited Members

4.68 LoadBalancingTransport Class Reference

Extends [LoadBalancingClient](#) with audio streaming functionality.

Inherits [LoadBalancingClient](#), [IVoiceTransport](#), and [IDisposable](#).

Inherited by [LoadBalancingFrontend](#), and [LoadBalancingTransport2](#).

Public Member Functions

- void **LogError** (string fmt, params object[] args)
- void **LogWarning** (string fmt, params object[] args)
- void **LogInfo** (string fmt, params object[] args)
- void **LogDebug** (string fmt, params object[] args)
- bool **IsChannelJoined** (int channelId)

- [LoadBalancingTransport](#) (ConnectionProtocol connectionProtocol=ConnectionProtocol.Udp)
Initializes a new [LoadBalancingTransport](#).
- new void [Service](#) ()
This method dispatches all available incoming commands and then sends this client's outgoing commands. Call this method regularly (2 to 20 times a second).
- virtual bool **ChangeAudioGroups** (byte[] groupsToRemove, byte[] groupsToAdd)
- void **SendVoicesInfo** (IEnumerable< [LocalVoice](#) > voices, int channelId, int targetPlayerId)
- void **SendVoiceRemove** ([LocalVoice](#) voice, int channelId, int targetPlayerId)
- virtual void **SendFrame** (ArraySegment< byte > data, FrameFlags flags, byte evNumber, byte voiceId, int channelId, int targetPlayerId, bool reliable, [LocalVoice](#) localVoice)
- string **ChannelIdStr** (int channelId)
- string **PlayerIdStr** (int playerId)
- void [Dispose](#) ()
Releases all resources used by the [LoadBalancingTransport](#) instance.

Protected Member Functions

- virtual void **onEventActionVoiceClient** (EventData ev)

Protected Attributes

- [VoiceClient](#) **voiceClient**

Properties

- [VoiceClient](#) **VoiceClient** [get]
The [VoiceClient](#) implementation associated with this [LoadBalancingTransport](#).
- byte **GlobalAudioGroup** [get, set]
- byte [GlobalInterestGroup](#) [get, set]
Set global audio group for this client. This call sets InterestGroup for existing local voices and for created later to given value. Client set as listening to this group only until LoadBalancingPeer.OpChangeGroups() called. This method can be called any time.

4.68.1 Detailed Description

Extends LoadBalancingClient with audio streaming functionality.

Use your normal LoadBalancing workflow to join a [Voice](#) room. All standard LoadBalancing features are available.

To work with audio:

- Create outgoing audio streams with Client.CreateLocalVoice.
- Handle new incoming audio streams info with OnRemoteVoiceInfoAction .
- Handle incoming audio streams data with OnAudioFrameAction .
- Handle closing of incoming audio streams with .

4.68.2 Constructor & Destructor Documentation

4.68.2.1 LoadBalancingTransport()

```
LoadBalancingTransport (
    ConnectionProtocol connectionProtocol = ConnectionProtocol.Udp )
```

Initializes a new [LoadBalancingTransport](#).

Parameters

<i>connectionProtocol</i>	Connection protocol (UDP or TCP). ConnectionProtocol
---------------------------	--

4.68.3 Member Function Documentation

4.68.3.1 Dispose()

```
void Dispose ( )
```

Releases all resources used by the [LoadBalancingTransport](#) instance.

4.68.3.2 Service()

```
new void Service ( )
```

This method dispatches all available incoming commands and then sends this client's outgoing commands. Call this method regularly (2 to 20 times a second).

4.68.4 Property Documentation

4.68.4.1 GlobalInterestGroup

```
byte GlobalInterestGroup [get], [set]
```

Set global audio group for this client. This call sets InterestGroup for existing local voices and for created later to given value. Client set as listening to this group only until LoadBalancingPeer.OpChangeGroups() called. This method can be called any time.

[LocalVoice.InterestGroup](#) LoadBalancingPeer.OpChangeGroups(byte[], byte[])

4.68.4.2 VoiceClient

```
VoiceClient VoiceClient [get]
```

The [VoiceClient](#) implementation associated with this [LoadBalancingTransport](#).

4.69 LoadBalancingTransport2 Class Reference

Variant of [LoadBalancingTransport](#). Aims to be non-alloc at the cost of breaking compatibility with older clients.

Inherits [LoadBalancingTransport](#).

Public Member Functions

- **LoadBalancingTransport2** (ConnectionProtocol connectionProtocol=ConnectionProtocol.Udp)
- override void **SendFrame** (ArraySegment< byte > data, FrameFlags flags, byte evNumber, byte voiceld, int channelId, int targetPlayerId, bool reliable, [LocalVoice](#) localVoice)

Protected Member Functions

- override void **onEventActionVoiceClient** (EventData ev)

Additional Inherited Members

4.69.1 Detailed Description

Variant of [LoadBalancingTransport](#). Aims to be non-alloc at the cost of breaking compatibility with older clients.

4.70 LocalVoice Class Reference

Represents outgoing data stream.

Inherits `IDisposable`.

Inherited by [LocalVoiceAudioDummy](#), and [LocalVoiceFramedBase](#).

Public Member Functions

- void [RemoveSelf](#) ()
Remove this voice from it's [VoiceClient](#) (using [VoiceClient.RemoveLocalVoice](#))
- virtual void **Dispose** ()

Static Public Attributes

- const int **DATA_POOL_CAPACITY** = 50

Protected Member Functions

- void **resetNoTransmitCnt** ()

Protected Attributes

- [VoiceInfo](#) **info**
- [IEncoder](#) **encoder**
- [VoiceClient](#) **voiceClient**
- `ArraySegment< byte >` **configFrame**
- volatile bool **disposed**
- object **disposeLock** = new object()

Properties

- byte **Group** [get, set]
- byte [InterestGroup](#) [get, set]
If InterestGroup != 0, voice's data is sent only to clients listening to this group (if supported by transport).
- [VoiceInfo](#) **Info** [get]
Returns Info structure assigned on local voice cration.
- bool [TransmitEnabled](#) [get, set]
If true, stream data broadcasted.
- bool [IsCurrentlyTransmitting](#) [get, protected set]
Returns true if stream broadcasts.
- int [FramesSent](#) [get]
Sent frames counter.
- int [FramesSentBytes](#) [get]
Sent frames bytes counter.
- bool [Reliable](#) [get, set]
Send data reliable.
- bool [Encrypt](#) [get, set]
Send data encrypted.
- [IServiceable](#) [LocalUserServiceable](#) [get, set]
Optional user object attached to [LocalVoice](#). its Service() will be called at each [VoiceClient.Service\(\)](#) call.
- bool [DebugEchoMode](#) [get, set]
If true, outgoing stream routed back to client via server same way as for remote client's streams. Can be swithed any time. OnRemoteVoiceInfoAction and OnRemoteVoiceRemoveAction are triggered if required. This functionality availability depends on transport.

4.70.1 Detailed Description

Represents outgoing data stream.

4.70.2 Member Function Documentation

4.70.2.1 RemoveSelf()

```
void RemoveSelf ( )
```

Remove this voice from it's [VoiceClient](#) (using [VoiceClient.RemoveLocalVoice](#)

.

4.70.3 Property Documentation

4.70.3.1 DebugEchoMode

```
bool DebugEchoMode [get], [set]
```

If true, outgoing stream routed back to client via server same way as for remote client's streams. Can be swithed any time. OnRemoteVoiceInfoAction and OnRemoteVoiceRemoveAction are triggered if required. This functionality availability depends on transport.

4.70.3.2 Encrypt

```
bool Encrypt [get], [set]
```

Send data encrypted.

4.70.3.3 FramesSent

```
int FramesSent [get]
```

Sent frames counter.

4.70.3.4 FramesSentBytes

```
int FramesSentBytes [get]
```

Sent frames bytes counter.

4.70.3.5 Info

```
VoiceInfo Info [get]
```

Returns Info structure assigned on local voice cration.

4.70.3.6 InterestGroup

```
byte InterestGroup [get], [set]
```

If InterestGroup != 0, voice's data is sent only to clients listening to this group (if supported by transport).

4.70.3.7 IsCurrentlyTransmitting

```
bool IsCurrentlyTransmitting [get], [protected set]
```

Returns true if stream broadcasts.

4.70.3.8 LocalUserServiceable

```
IServiceable LocalUserServiceable [get], [set]
```

Optional user object attached to [LocalVoice](#). its Service() will be called at each [VoiceClient.Service\(\)](#) call.

4.70.3.9 Reliable

```
bool Reliable [get], [set]
```

Send data reliable.

4.70.3.10 TransmitEnabled

```
bool TransmitEnabled [get], [set]
```

If true, stream data broadcasted.

4.71 LocalVoiceAudio< T > Class Template Reference

Outgoing audio stream.

Inherits [LocalVoiceFramed< T >](#), and [ILocalVoiceAudio](#).

Public Member Functions

- void [VoiceDetectorCalibrate](#) (int durationMs, Action< float > onCalibrated=null)
Trigger voice detector calibration process.

Static Public Member Functions

- static [LocalVoiceAudio](#)< T > [Create](#) ([VoiceClient](#) voiceClient, byte voiceId, [IEncoder](#) encoder, [VoiceInfo](#) voiceInfo, [IAudioDesc](#) audioSourceDesc, int channelId)

Create a new [LocalVoiceAudio](#)<T> instance.

Protected Member Functions

- void [initBuiltinProcessors](#) ()

Protected Attributes

- [AudioUtil.VoiceDetector](#)< T > **voiceDetector**
- [AudioUtil.VoiceDetectorCalibration](#)< T > **voiceDetectorCalibration**
- [AudioUtil.LevelMeter](#)< T > **levelMeter**
- int **channels**
- bool **resampleSource**

Properties

- virtual [AudioUtil.IVoiceDetector](#) **VoiceDetector** [get]
- virtual [AudioUtil.ILevelMeter](#) **LevelMeter** [get]
- bool [VoiceDetectorCalibrating](#) [get]

True if the [VoiceDetector](#) is currently calibrating.

Additional Inherited Members

4.71.1 Detailed Description

Outgoing audio stream.

4.71.2 Member Function Documentation

4.71.2.1 [Create\(\)](#)

```
static LocalVoiceAudio<T> Create (
    VoiceClient voiceClient,
    byte voiceId,
    IEncoder encoder,
    VoiceInfo voiceInfo,
    IAudioDesc audioSourceDesc,
    int channelId ) [static]
```

Create a new [LocalVoiceAudio](#)<T> instance.

Parameters

<i>voiceClient</i>	The VoiceClient to use for this outgoing stream.
<i>voiceId</i>	Numeric ID for this voice.
<i>encoder</i>	Encoder to use for this voice.
<i>channelId</i>	Voice transport channel ID to use for this voice.

Returns

The new LocalVoiceAudio<T> instance.

4.71.2.2 VoiceDetectorCalibrate()

```
void VoiceDetectorCalibrate (
    int durationMs,
    Action< float > onCalibrated = null )
```

Trigger voice detector calibration process.

While calibrating, keep silence. [Voice](#) detector sets threshold basing on measured background noise level.

Parameters

<i>durationMs</i>	Duration of calibration in milliseconds.
<i>onCalibrated</i>	Called when calibration is complete. Parameter is new threshold value.

Implements [ILocalVoiceAudio](#).

4.71.3 Property Documentation

4.71.3.1 VoiceDetectorCalibrating

```
bool VoiceDetectorCalibrating [get]
```

True if the VoiceDetector is currently calibrating.

4.72 LocalVoiceAudioDummy Class Reference

Dummy [LocalVoiceAudio](#)

Inherits [LocalVoice](#), and [ILocalVoiceAudio](#).

Public Member Functions

- void [VoiceDetectorCalibrate](#) (int durationMs, Action< float > onCalibrated=null)
Trigger voice detector calibration process.

Static Public Attributes

- static [LocalVoiceAudioDummy Dummy](#) = new [LocalVoiceAudioDummy](#)()
A Dummy [LocalVoiceAudio](#) instance.

Properties

- [AudioUtil.IVoiceDetector VoiceDetector](#) [get]
- [AudioUtil.ILevelMeter LevelMeter](#) [get]
- bool [VoiceDetectorCalibrating](#) [get]

Additional Inherited Members

4.72.1 Detailed Description

Dummy [LocalVoiceAudio](#)

For testing, this [LocalVoiceAudio](#) implementation features a [AudioUtil.VoiceDetectorDummy](#) and a [AudioUtil.LevelMeterDummy](#)

4.72.2 Member Function Documentation

4.72.2.1 VoiceDetectorCalibrate()

```
void VoiceDetectorCalibrate (  
    int durationMs,  
    Action< float > onCalibrated = null )
```

Trigger voice detector calibration process.

While calibrating, keep silence. [Voice](#) detector sets threshold based on measured background noise level.

Parameters

<i>durationMs</i>	Duration of calibration (in milliseconds).
<i>onCalibrated</i>	Called when calibration is complete. Parameter is new threshold value.

Implements [ILocalVoiceAudio](#).

4.72.3 Member Data Documentation

4.72.3.1 Dummy

```
LocalVoiceAudioDummy Dummy = new LocalVoiceAudioDummy() [static]
```

A Dummy [LocalVoiceAudio](#) instance.

4.73 LocalVoiceAudioFloat Class Reference

Specialization of [LocalVoiceAudio](#) for float audio

Inherits [LocalVoiceAudio< float >](#).

Additional Inherited Members

4.73.1 Detailed Description

Specialization of [LocalVoiceAudio](#) for float audio

4.74 LocalVoiceAudioShort Class Reference

Specialization of [LocalVoiceAudio](#) for short audio

Inherits [LocalVoiceAudio< short >](#).

Additional Inherited Members

4.74.1 Detailed Description

Specialization of [LocalVoiceAudio](#) for short audio

4.75 LocalVoiceFramed< T > Class Template Reference

Typed re-framing [LocalVoice](#)

Inherits [LocalVoiceFramedBase](#).

Inherited by [LocalVoiceAudio< T >](#).

Public Member Functions

- void [AddPostProcessor](#) (params [IProcessor](#)< T >[] processors)
Adds processors after any built-in processors and everything added with [AddPreProcessor](#).
- void [AddPreProcessor](#) (params [IProcessor](#)< T >[] processors)
Adds processors before built-in processors and everything added with [AddPostProcessor](#).
- void [ClearProcessors](#) ()
Clears all processors in pipeline including built-in resampling. User should add at least resampler processor after call.
- void [PushDataAsync](#) (T[] buf)
Asynchronously push data into this stream.
- void [PushData](#) (T[] buf)
Synchronously push data into this stream.
- override void [Dispose](#) ()
Releases resources used by the [VoiceFramed](#) instance. Buffers used for asynchronous push will be disposed in encoder thread's 'finally'.

Protected Member Functions

- T[] [processFrame](#) (T[] buf)

Properties

- [FactoryPrimitiveArrayPool](#)< T > **BufferFactory** [get]
- bool [PushDataAsyncReady](#) [get]
Wether this [LocalVoiceFramed](#) has capacity for more data buffers to be pushed asynchronously.

Additional Inherited Members

4.75.1 Detailed Description

Typed re-framing [LocalVoice](#)

Consumes data in array buffers of arbitrary length. Repacks them in frames of constant length for further processing and encoding.

Parameters

<i>voiceInfo</i>	Outgoing stream parameters. Set applicable fields to read them by encoder and by receiving client when voice created.
<i>channel↔ Id</i>	Transport channel specific to transport.
<i>encoder</i>	Encoder producing the stream.

Returns

Outgoing stream handler.

4.75.2 Member Function Documentation

4.75.2.1 AddPostProcessor()

```
void AddPostProcessor (
    params IProcessor< T >[] processors )
```

Adds processors after any built-in processors and everything added with AddPreProcessor.

Parameters

<i>processors</i>	
-------------------	--

4.75.2.2 AddPreProcessor()

```
void AddPreProcessor (
    params IProcessor< T >[] processors )
```

Adds processors before built-in processors and everything added with AddPostProcessor.

Parameters

<i>processors</i>	
-------------------	--

4.75.2.3 ClearProcessors()

```
void ClearProcessors ( )
```

Clears all processors in pipeline including built-in resampling. User should add at least resampler processor after call.

4.75.2.4 Dispose()

```
override void Dispose ( ) [virtual]
```

Releases resources used by the VoiceFramed instance. Buffers used for asynchronous push will be disposed in encoder thread's 'finally'.

Reimplemented from [LocalVoice](#).

4.75.2.5 PushData()

```
void PushData (
    T[] buf )
```

Synchronously push data into this stream.

4.75.2.6 PushDataAsync()

```
void PushDataAsync (
    T[] buf )
```

Asynchronously push data into this stream.

4.75.3 Property Documentation

4.75.3.1 PushDataAsyncReady

```
bool PushDataAsyncReady [get]
```

Whether this [LocalVoiceFramed](#) has capacity for more data buffers to be pushed asynchronously.

4.76 LocalVoiceFramedBase Class Reference

Typed re-framing [LocalVoice](#)

Inherits [LocalVoice](#).

Inherited by [LocalVoiceFramed< T >](#).

Properties

- int [FrameSize](#) [get]

Data flow will be repacked to frames of this size. May differ from input voiceInfo.FrameSize. Processors should resample in this case.

Additional Inherited Members

4.76.1 Detailed Description

Typed re-framing [LocalVoice](#)

Base class for typed re-framing [LocalVoice](#) implementation ([LocalVoiceFramedBase<T>](#))

4.76.2 Property Documentation

4.76.2.1 FrameSize

int FrameSize [get]

Data flow will be repacked to frames of this size. May differ from input voiceInfo.FrameSize. Processors should resample in this case.

4.77 Logger Class Reference

Inherits [ILogger](#).

Public Member Functions

- void **LogError** (string fmt, params object[] args)
- void **LogWarning** (string fmt, params object[] args)
- void **LogInfo** (string fmt, params object[] args)
- void **LogDebug** (string fmt, params object[] args)

4.78 MicAmplifier Class Reference

Inherits [VoiceComponent](#).

Properties

- float **AmplificationFactor** [get, set]
- float **BoostValue** [get, set]

Additional Inherited Members

4.79 MicAmplifierFloat Class Reference

Inherits [IProcessor< float >](#).

Public Member Functions

- **MicAmplifierFloat** (float amplificationFactor, float boostValue)
- float[] **Process** (float[] buf)
- void **Dispose** ()

Properties

- float **AmplificationFactor** [get, set]
- float **BoostValue** [get, set]
- float **MaxBefore** [get]
- float **MaxAfter** [get]
- bool **Disabled** [get, set]

4.80 MicAmplifierShort Class Reference

Inherits [IProcessor< short >](#).

Public Member Functions

- **MicAmplifierShort** (short amplificationFactor, short boostValue)
- short[] **Process** (short[] buf)
- void **Dispose** ()

Properties

- short **AmplificationFactor** [get, set]
- short **BoostValue** [get, set]
- short **MaxBefore** [get]
- short **MaxAfter** [get]
- bool **Disabled** [get, set]

4.81 MicWrapper Class Reference

Inherits [IAudioReader< float >](#).

Public Member Functions

- **MicWrapper** (string device, int suggestedFrequency, [ILogger](#) logger)
- void **Dispose** ()
- bool **Read** (float[] buffer)

Properties

- int? **SamplingRate** [get]
- int? **Channels** [get]
- string **Error** [get]

4.82 MicWrapperPusher Class Reference

Inherits [IAudioPusher< float >](#).

Public Member Functions

- **MicWrapperPusher** (string device, AudioSource aS, int suggestedFrequency, [ILogger](#) lg, bool destroyOn↵Stop=true)
- **MicWrapperPusher** (string device, GameObject gO, int suggestedFrequency, [ILogger](#) lg, bool destroyOn↵Stop=true)
- **MicWrapperPusher** (string device, Transform parentTransform, int suggestedFrequency, [ILogger](#) lg, bool destroyOnStop=true)
- void **SetCallback** (Action< float[]> callback, [ObjectFactory](#)< float[], int > bufferFactory)
- void **Dispose** ()

Properties

- int? **SamplingRate** [get]
- int? **Channels** [get]
- string **Error** [get]

4.83 MonoPInvokeCallbackAttribute Class Reference

Inherits Attribute.

Public Member Functions

- **MonoPInvokeCallbackAttribute** (Type t)

4.84 NativeAndroidMicrophoneSettings Struct Reference

Public Attributes

- bool **AcousticEchoCancellation**
- bool **AutomaticGainControl**
- bool **NoiseSuppression**

4.85 ObjectFactory< TType, TInfo > Interface Template Reference

Uniform interface to ObjectPool<TType, TInfo> and single reusable object.

Inherits IDisposable.

Public Member Functions

- TType **New** ()
- TType **New** (TInfo info)
- void **Free** (TType obj)
- void **Free** (TType obj, TInfo info)

Properties

- TInfo **Info** [get]

4.85.1 Detailed Description

Uniform interface to ObjectPool<TType, TInfo> and single reusable object.

Template Parameters

<i>TType</i>	Object type.
<i>TInfo</i>	Type of property used to check 2 objects identity (like integral length of array).

4.86 ObjectPool< TType, TInfo > Class Template Reference

Generic Pool to re-use objects of a certain type (TType) that optionally match a certain property or set of properties (TInfo).

Inherits IDisposable.

Public Member Functions

- [ObjectPool](#) (int capacity, string name)
Create a new [ObjectPool](#) instance. Does not call [Init\(\)](#).
- [ObjectPool](#) (int capacity, string name, TInfo info)
Create a new [ObjectPool](#) instance with the given info structure. Calls [Init\(\)](#).
- void [Init](#) (TInfo info)
(Re-)Initializes this [ObjectPool](#).
- TType [AcquireOrCreate](#) ()
Acquire an existing object, or create a new one if none are available.
- TType [AcquireOrCreate](#) (TInfo info)
Acquire an existing object (if info matches), or create a new one from the passed info.
- virtual bool [Release](#) (TType obj, TInfo objInfo)
Returns object to pool.
- virtual bool [Release](#) (TType obj)
Returns object to pool, or destroys it if the pool is full.
- void [Dispose](#) ()
Free resources associated with this [ObjectPool](#)

Protected Member Functions

- abstract TType **createObject** (TInfo info)
- abstract void **destroyObject** (TType obj)
- abstract bool **infosMatch** (TInfo i0, TInfo i1)

Protected Attributes

- int **capacity**
- TInfo **info**
- int **pos**
- string **name**

Properties

- TInfo [Info](#) [get]
The property (info) that objects in this Pool must match.

4.86.1 Detailed Description

Generic Pool to re-use objects of a certain type (TType) that optionally match a certain property or set of properties (TInfo).

Template Parameters

<i>TType</i>	Object type.
<i>TInfo</i>	Type of parameter used to check 2 objects identity (like integral length of array).

4.86.2 Constructor & Destructor Documentation

4.86.2.1 ObjectPool() [1/2]

```
ObjectPool (
    int capacity,
    string name )
```

Create a new [ObjectPool](#) instance. Does not call [Init\(\)](#).

Parameters

<i>capacity</i>	Capacity (size) of the object pool.
<i>name</i>	Name of the object pool.

4.86.2.2 ObjectPool() [2/2]

```
ObjectPool (
    int capacity,
    string name,
    TInfo info )
```

Create a new [ObjectPool](#) instance with the given info structure. Calls [Init\(\)](#).

Parameters

<i>capacity</i>	Capacity (size) of the object pool.
<i>name</i>	Name of the object pool.
<i>info</i>	Info about this Pool's objects.

4.86.3 Member Function Documentation

4.86.3.1 AcquireOrCreate() [1/2]

```
TType AcquireOrCreate ( )
```

Acquire an existing object, or create a new one if none are available.

If it fails to get one from the pool, this will create from the info given in this pool's constructor.

4.86.3.2 AcquireOrCreate() [2/2]

```
TType AcquireOrCreate (
    TInfo info )
```

Acquire an existing object (if info matches), or create a new one from the passed info.

Parameters

<i>info</i>	Info structure to match, or create a new object with.
-------------	---

4.86.3.3 Dispose()

```
void Dispose ( )
```

Free resources associated with this [ObjectPool](#)

4.86.3.4 Init()

```
void Init (
    TInfo info )
```

(Re-)Initializes this [ObjectPool](#).

If there are objects available in this Pool, they will be destroyed. Allocates (Capacity) new Objects.

Parameters

<i>info</i>	Info about this Pool's objects.
-------------	---------------------------------

4.86.3.5 Release() [1/2]

```
virtual bool Release (
    TType obj ) [virtual]
```

Returns object to pool, or destroys it if the pool is full.

Parameters

<i>obj</i>	The object to return to the pool.
------------	-----------------------------------

4.86.3.6 Release() [2/2]

```
virtual bool Release (
    TType obj,
    TInfo objInfo ) [virtual]
```

Returns object to pool.

Parameters

<i>obj</i>	The object to return to the pool.
<i>objInfo</i>	The info structure about obj.

obj is returned to the pool only if *objInfo* matches this pool's info. Else, it is destroyed.

4.86.4 Property Documentation

4.86.4.1 Info

```
TInfo Info [get]
```

The property (info) that objects in this Pool must match.

4.87 OpusCodec Class Reference

Classes

- class [Decoder](#)
- class [DecoderFactory](#)
- class [Encoder](#)
- class [EncoderFloat](#)
- class [EncoderShort](#)
- class [Factory](#)
- class [Util](#)

Public Types

- enum **FrameDuration**

Properties

- static string **Version** [get]

4.88 OpusDecoder< T > Class Template Reference

Inherits IDisposable.

Public Member Functions

- **OpusDecoder** (SamplingRate outputSamplingRateHz, [Channels](#) numChannels)
- T[] **DecodePacket** (byte[] packetData)
- T[] **DecodeEndOfStream** ()
- void **Dispose** ()

Properties

- [Bandwidth](#)? **PreviousPacketBandwidth** [get]

4.89 OpusEncoder Class Reference

Inherits IDisposable.

Public Member Functions

- **OpusEncoder** (SamplingRate inputSamplingRateHz, [Channels](#) numChannels, int bitrate, [OpusApplicationType](#) applicationType, [Delay](#) encoderDelay)
- ArraySegment< byte > **Encode** (float[] pcmSamples)
- ArraySegment< byte > **Encode** (short[] pcmSamples)
- void **Dispose** ()

Static Public Attributes

- const int **BitrateMax** = -1

Properties

- SamplingRate **InputSamplingRate** [get]
- Channels **InputChannels** [get]
- Delay **EncoderDelay** [get, set]
Using a duration of less than 10 ms will prevent the encoder from using the LPC or hybrid modes.
- int **FrameSizePerChannel** [get]
- int **Bitrate** [get, set]
- Bandwidth **MaxBandwidth** [get, set]
- Complexity **Complexity** [get, set]
- int **ExpectedPacketLossPercentage** [get, set]
- SignalHint **SignalHint** [get, set]
- ForceChannels **ForceChannels** [get, set]
- bool? **UseInbandFEC** [get, set]
- int **PacketLossPercentage** [get, set]
- bool? **UseUnconstrainedVBR** [get, set]
- bool? **DtxEnabled** [get, set]

4.89.1 Property Documentation

4.89.1.1 EncoderDelay

Delay **EncoderDelay** [get], [set]

Using a duration of less than 10 ms will prevent the encoder from using the LPC or hybrid modes.

4.90 OpusException Class Reference

Inherits Exception.

Public Member Functions

- **OpusException** (OpusStatusCode statusCode, string message)

Properties

- OpusStatusCode **Statuscode** [get]

4.91 OpusLib Class Reference

Properties

- static string **Version** [get]

4.92 Recorder.PhotonVoiceCreatedParams Class Reference

Inherits [PhotonVoiceCreatedParams](#).

Additional Inherited Members

4.93 PhotonVoiceCreatedParams Class Reference

Inherited by [Recorder.PhotonVoiceCreatedParams](#).

Properties

- [Voice.LocalVoice](#) **Voice** [get, set]
- [Voice.IAudioDesc](#) **AudioDesc** [get, set]

4.94 PhotonVoiceLagSimulationGui Class Reference

Inherits [MonoBehaviour](#).

Public Member Functions

- void **OnEnable** ()

4.95 PhotonVoiceNetwork Class Reference

This class can be used to automatically sync client states between [PUN](#) and [Voice](#). It also sets a custom [PUN](#) Speaker factory to find the Speaker component for a character's voice. For this to work attach a [PhotonVoiceView](#) next to the PhotonView of your player's prefab.

Inherits [VoiceConnection](#).

Public Member Functions

- bool [ConnectAndJoinRoom](#) ()
Connect voice client to [Photon](#) servers and join a [Voice](#) room
- void [Disconnect](#) ()
Disconnect voice client from all [Photon](#) servers

Public Attributes

- bool [AutoConnectAndJoin](#) = true
Auto connect voice client and join a voice room when [PUN](#) client is joined to a [PUN](#) room
- bool [AutoLeaveAndDisconnect](#) = true
Auto disconnect voice client when [PUN](#) client is not joined to a [PUN](#) room
- bool [WorkInOfflineMode](#) = true
Whether or not [Photon Voice](#) client should follow [PUN](#) client if the latter is in offline mode.

Static Public Attributes

- const string [VoiceRoomNameSuffix](#) = "__voice_"
Suffix for voice room names appended to [PUN](#) room names.

Protected Member Functions

- override void **Awake** ()
- override void **OnDisable** ()
- override void **OnDestroy** ()
- override void **OnVoiceStateChanged** ([ClientState](#) fromState, [ClientState](#) toState)
- override [Speaker](#) **SimpleSpeakerFactory** (int playerId, byte voiceId, object userData)

Properties

- static [PhotonVoiceNetwork](#) [Instance](#) [get, set]
Singleton instance for [PhotonVoiceNetwork](#)
- bool [UsePunAuthValues](#) [get, set]
Whether or not to use the same [PhotonNetwork.AuthValues](#) in [PhotonVoiceNetwork.Instance.Client.AuthValues](#).

Additional Inherited Members

4.95.1 Detailed Description

This class can be used to automatically sync client states between [PUN](#) and [Voice](#). It also sets a custom [PUN](#) Speaker factory to find the Speaker component for a character's voice. For this to work attach a [PhotonVoiceView](#) next to the PhotonView of your player's prefab.

4.95.2 Member Function Documentation

4.95.2.1 ConnectAndJoinRoom()

```
bool ConnectAndJoinRoom ( )
```

Connect voice client to [Photon](#) servers and join a [Voice](#) room

Returns

If true, connection command send from client

4.95.2.2 Disconnect()

```
void Disconnect ( )
```

Disconnect voice client from all [Photon](#) servers

4.95.3 Member Data Documentation

4.95.3.1 AutoConnectAndJoin

```
bool AutoConnectAndJoin = true
```

Auto connect voice client and join a voice room when [PUN](#) client is joined to a [PUN](#) room

4.95.3.2 AutoLeaveAndDisconnect

```
bool AutoLeaveAndDisconnect = true
```

Auto disconnect voice client when [PUN](#) client is not joined to a [PUN](#) room

4.95.3.3 VoiceRoomNameSuffix

```
const string VoiceRoomNameSuffix = "_voice_" [static]
```

Suffix for voice room names appended to [PUN](#) room names.

4.95.3.4 WorkInOfflineMode

```
bool WorkInOfflineMode = true
```

Whether or not [Photon Voice](#) client should follow [PUN](#) client if the latter is in offline mode.

4.95.4 Property Documentation

4.95.4.1 Instance

```
PhotonVoiceNetwork Instance [static], [get], [set]
```

Singleton instance for [PhotonVoiceNetwork](#)

4.95.4.2 UsePunAuthValues

```
bool UsePunAuthValues [get], [set]
```

Whether or not to use the same [PhotonNetwork.AuthValues](#) in [PhotonVoiceNetwork.Instance.Client.AuthValues](#).

4.96 PhotonVoiceStatsGui Class Reference

Basic GUI to show traffic and health statistics of the connection to [Photon](#), toggled by shift+tab.

Inherits [MonoBehaviour](#).

4.96.1 Detailed Description

Basic GUI to show traffic and health statistics of the connection to [Photon](#), toggled by shift+tab.

The shown health values can help identify problems with connection losses or performance. Example: If the time delta between two consecutive `SendOutgoingCommands` calls is a second or more, chances rise for a disconnect being caused by this (because acknowledgments to the server need to be sent in due time).

4.97 PhotonVoiceView Class Reference

Component that should be attached to a networked [PUN](#) prefab that has [PhotonView](#). It will bind remote [Recorder](#) with local [Speaker](#) of the same networked prefab. This component makes automatic voice stream routing easy for players' characters/avatars.

Inherits [VoiceComponent](#).

Public Member Functions

- void [Init](#) ()
Initializes this [PhotonVoiceView](#) for [Voice](#) usage based on the [PhotonView](#), [Recorder](#) and [Speaker](#) components.

Public Attributes

- bool [AutoCreateRecorderIfNotFound](#)
If true, a [Recorder](#) component will be added to the same [GameObject](#) if not found already.
- bool [UsePrimaryRecorder](#)
If true, [PhotonVoiceNetwork.PrimaryRecorder](#) will be used by this [PhotonVoiceView](#)
- bool [SetupDebugSpeaker](#)
If true, a [Speaker](#) component will be setup to be used for the [DebugEcho](#) mode

Protected Member Functions

- override void **Awake** ()

Properties

- [Recorder](#) [RecorderInUse](#) [get, set]
The [Recorder](#) component currently used by this [PhotonVoiceView](#)
- [Speaker](#) [SpeakerInUse](#) [get, set]
The [Speaker](#) component currently used by this [PhotonVoiceView](#)
- bool [IsSetup](#) [get]
If true, this [PhotonVoiceView](#) is setup and ready to be used
- bool [IsSpeaker](#) [get]
If true, this [PhotonVoiceView](#) has a [Speaker](#) setup for playback of received audio frames from remote audio source
- bool [IsSpeaking](#) [get]
If true, this [PhotonVoiceView](#) has a [Speaker](#) that is currently playing received audio frames from remote audio source
- bool [IsRecorder](#) [get]
If true, this [PhotonVoiceView](#) has a [Recorder](#) setup for transmission of audio stream from local audio source
- bool [IsRecording](#) [get]
If true, this [PhotonVoiceView](#) has a [Recorder](#) that is currently transmitting audio stream from local audio source
- bool [IsSpeakerLinked](#) [get]
If true, the [SpeakerInUse](#) is linked to the remote voice stream
- bool [IsPhotonViewReady](#) [get]
If true, the [PhotonView](#) attached to the same [GameObject](#) has a valid [ViewID](#) > 0

Additional Inherited Members

4.97.1 Detailed Description

Component that should be attached to a networked [PUN](#) prefab that has [PhotonView](#). It will bind remote [Recorder](#) with local [Speaker](#) of the same networked prefab. This component makes automatic voice stream routing easy for players' characters/avatars.

4.97.2 Member Function Documentation

4.97.2.1 Init()

```
void Init ( )
```

Initializes this [PhotonVoiceView](#) for [Voice](#) usage based on the PhotonView, Recorder and Speaker components.

The initialization should happen automatically. Call this method explicitly if this does not succeed. The initialization is a two steps operation: step one is the setup of Recorder and Speaker to be used. Step two is the late-linking -if needed- of the SpeakerInUse and corresponding remote voice info -if any- via ViewID.

4.97.3 Member Data Documentation

4.97.3.1 AutoCreateRecorderIfNotFound

```
bool AutoCreateRecorderIfNotFound
```

If true, a Recorder component will be added to the same GameObject if not found already.

4.97.3.2 SetupDebugSpeaker

```
bool SetupDebugSpeaker
```

If true, a Speaker component will be setup to be used for the DebugEcho mode

4.97.3.3 UsePrimaryRecorder

```
bool UsePrimaryRecorder
```

If true, [PhotonVoiceNetwork.PrimaryRecorder](#) will be used by this [PhotonVoiceView](#)

4.97.4 Property Documentation

4.97.4.1 IsPhotonViewReady

```
bool IsPhotonViewReady [get]
```

If true, the PhotonView attached to the same GameObject has a valid ViewID > 0

4.97.4.2 IsRecorder

```
bool IsRecorder [get]
```

If true, this [PhotonVoiceView](#) has a Recorder setup for transmission of audio stream from local audio source

4.97.4.3 IsRecording

```
bool IsRecording [get]
```

If true, this [PhotonVoiceView](#) has a Recorder that is currently transmitting audio stream from local audio source

4.97.4.4 IsSetup

```
bool IsSetup [get]
```

If true, this [PhotonVoiceView](#) is setup and ready to be used

4.97.4.5 IsSpeaker

```
bool IsSpeaker [get]
```

If true, this [PhotonVoiceView](#) has a Speaker setup for playback of received audio frames from remote audio source

4.97.4.6 IsSpeakerLinked

```
bool IsSpeakerLinked [get]
```

If true, the SpeakerInUse is linked to the remote voice stream

4.97.4.7 IsSpeaking

```
bool IsSpeaking [get]
```

If true, this [PhotonVoiceView](#) has a Speaker that is currently playing received audio frames from remote audio source

4.97.4.8 RecorderInUse

```
Recorder RecorderInUse [get], [set]
```

The Recorder component currently used by this [PhotonVoiceView](#)

4.97.4.9 SpeakerInUse

```
Speaker SpeakerInUse [get], [set]
```

The Speaker component currently used by this [PhotonVoiceView](#)

4.98 Platform Class Reference

Static Public Member Functions

- static [IEncoder](#) [CreateDefaultAudioEncoder](#)< T > ([ILogger](#) logger, [VoiceInfo](#) info)

4.99 PlaybackDelaySettings Struct Reference

Playback delay configuration container.

Public Member Functions

- override string [ToString](#) ()

Public Attributes

- int [MinDelaySoft](#)
ms: Audio player tries to keep the delay above this value.
- int [MaxDelaySoft](#)
ms: Audio player tries to keep the delay below this value.
- int [MaxDelayHard](#)
ms: Audio player guarantees that the delay never exceeds this value.

Static Public Attributes

- const int **DEFAULT_LOW** = 200
- const int **DEFAULT_HIGH** = 400
- const int **DEFAULT_MAX** = 1000

4.99.1 Detailed Description

Playback delay configuration container.

4.99.2 Member Data Documentation

4.99.2.1 MaxDelayHard

```
int MaxDelayHard
```

ms: Audio player guarantees that the delay never exceeds this value.

4.99.2.2 MaxDelaySoft

```
int MaxDelaySoft
```

ms: Audio player tries to keep the delay below this value.

4.99.2.3 MinDelaySoft

```
int MinDelaySoft
```

ms: Audio player tries to keep the delay above this value.

4.100 UnityAudioOut.PlayDelayConfig Struct Reference

Static Public Attributes

- static [PlayDelayConfig](#) **Default** = new [PlayDelayConfig](#) { Low = 200, High = 400, Max = 1000 }

Properties

- int **Low** [get, set]
- int **High** [get, set]
- int **Max** [get, set]

4.101 PrimitiveArrayPool< T > Class Template Reference

Pool of Arrays with components of type T, with [ObjectPool](#) info being the array's size.

Inherits [ObjectPool< T\[\], int >](#).

Public Member Functions

- **PrimitiveArrayPool** (int capacity, string name)
- **PrimitiveArrayPool** (int capacity, string name, int info)

Protected Member Functions

- override T[] **createObject** (int info)
- override void **destroyObject** (T[] obj)
- override bool **infosMatch** (int i0, int i1)

Additional Inherited Members

4.101.1 Detailed Description

Pool of Arrays with components of type T, with [ObjectPool](#) info being the array's size.

Template Parameters

<i>T</i>	Array element type.
----------	---------------------

4.102 RawCodec Class Reference

Classes

- class [Decoder](#)
- class [Encoder](#)

4.103 Recorder Class Reference

Component representing outgoing audio stream in scene.

Inherits [VoiceComponent](#).

Classes

- class [PhotonVoiceCreatedParams](#)

Public Types

- enum **InputSourceType**
- enum **MicType**
- enum **SampleTypeConv**

Public Member Functions

- void [Init](#) ([VoiceClient](#) voiceClient, object customObj=null)
Initializes the [Recorder](#) component to be able to transmit audio.
- void [Init](#) ([VoiceConnection](#) voiceConnection)
Initializes the [Recorder](#) component to be able to transmit audio.
- void **Relnit** ()
- void [RestartRecording](#) (bool force=false)
Restarts recording if something has changed that requires this.
- void [VoiceDetectorCalibrate](#) (int durationMs, Action< float > detectionEndedCallback=null)
Trigger voice detector calibration process. While calibrating, keep silence. [Voice](#) detector sets threshold basing on measured background noise level.
- void [StartRecording](#) ()
Starts recording.
- void [StopRecording](#) ()
Stops recording.
- bool [ResetLocalAudio](#) ()
Resets audio session and parameters locally to fix broken recording due to system configuration modifications or audio interruptions or audio routing changes.

Static Public Member Functions

- static bool **CompareUnityMicNames** (string mic1, string mic2)
- static bool **IsDefaultUnityMic** (string mic)

Static Public Attributes

- const int **MIN_OPUS_BITRATE** = 6000
- const int **MAX_OPUS_BITRATE** = 510000

Protected Member Functions

- virtual void **SendPhotonVoiceCreatedMessage** ()

Properties

- static [AudioInEnumerator PhotonMicrophoneEnumerator](#) [get]
Enumerator for the available microphone devices gathered by the [Photon](#) plugin.
- bool [IsInitialized](#) [get]
If true, this [Recorder](#) has been initialized and is ready to transmit to remote clients. Otherwise call [Init\(VoiceConnection\)](#).
- bool [RequiresInit](#) [get]
- bool [RequiresRestart](#) [get, protected set]
Returns true if something has changed in the [Recorder](#) while recording that won't take effect unless recording is restarted using [RestartRecording](#).
- bool [TransmitEnabled](#) [get, set]
If true, audio transmission is enabled.
- bool [Encrypt](#) [get, set]
If true, voice stream is sent encrypted.
- bool [DebugEchoMode](#) [get, set]
If true, outgoing stream routed back to client via server same way as for remote client's streams.
- bool [ReliableMode](#) [get, set]
If true, stream data sent in reliable mode.
- bool [VoiceDetection](#) [get, set]
If true, voice detection enabled.
- float [VoiceDetectionThreshold](#) [get, set]
Voice detection threshold (0..1, where 1 is full amplitude).
- int [VoiceDetectionDelayMs](#) [get, set]
Keep detected state during this time after signal level dropped below threshold. Default is 500ms
- object [UserData](#) [get, set]
Custom user object to be sent in the voice stream info event.
- Func< [IAudioDesc](#) > [InputFactory](#) [get, set]
Set the method returning new [Voice.IAudioDesc](#) instance to be assigned to a new voice created with Source set to Factory
- [AudioUtil.IVoiceDetector?](#) [VoiceDetector](#) [get]
Returns voice activity detector for recorder's audio stream.
- string [UnityMicrophoneDevice](#) [get, set]
Set or get [Unity](#) microphone device used for streaming.
- int [PhotonMicrophoneDeviceId](#) [get, set]
Set or get photon microphone device used for streaming.
- byte [AudioGroup](#) [get, set]
Target interest group that will receive transmitted audio.
- byte [InterestGroup](#) [get, set]
Target interest group that will receive transmitted audio.
- bool [IsCurrentlyTransmitting](#) [get]
Returns true if audio stream broadcasts.
- [AudioUtil.ILevelMeter?](#) [LevelMeter](#) [get]
Level meter utility.
- bool [VoiceDetectorCalibrating](#) [get]
If true, voice detector calibration is in progress.
- [ILocalVoiceAudio](#) [voiceAudio](#) [get]
- InputSourceType [SourceType](#) [get, set]
Audio data source.
- MicType [MicrophoneType](#) [get, set]
Which microphone API to use when the Source is set to Microphone.
- SampleTypeConv [TypeConvert](#) [get, set]

- Force creation of 'short' pipeline and convert audio data to short for 'float' audio sources.*

 - AudioClip [AudioClip](#) [get, set]
Source audio clip.
 - bool [LoopAudioClip](#) [get, set]
Loop playback for audio clip sources.
 - SamplingRate [SamplingRate](#) [get, set]
Outgoing audio stream sampling rate.
 - OpusCodec.FrameDuration [FrameDuration](#) [get, set]
Outgoing audio stream encoder delay.
 - int [Bitrate](#) [get, set]
Outgoing audio stream bitrate.
 - bool [IsRecording](#) [get, set]
Gets or sets whether this [Recorder](#) is actively recording audio to be transmitted.
 - bool [ReactOnSystemChanges](#) [get, set]
If true, the [Recorder](#) will automatically restart recording to recover from audio device changes.
 - bool [AutoStart](#) [get, set]
If true, automatically start recording when initialized.
 - bool [RecordOnlyWhenEnabled](#) [get, set]
If true, component will work only when enabled and active in hierarchy.
 - bool [SkipDeviceChangeChecks](#) [get, set]
If true, restarts recording without checking if audio config/device changes affected recording.
 - bool [StopRecordingWhenPaused](#) [get, set]
If true, stop recording when paused resume/restart when un-paused.
 - bool [UseOnAudioFilterRead](#) [get, set]
If true, recording will make use of [Unity](#)'s OnAudioFiltterRead callback from a muted local AudioSource.
 - bool [TrySamplingRateMatch](#) [get, set]
If true, [Recorder](#) will try to match sampling rates of microphone device and Opus encoder to avoid re sampling of audio input.
 - bool [UseMicrophoneTypeFallback](#) [get, set]
If true, if recording fails to start with [Unity](#) microphone type, [Photon](#) microphone type is used -if available- as a fallback and vice versa.

Additional Inherited Members

4.103.1 Detailed Description

Component representing outgoing audio stream in scene.

4.103.2 Member Function Documentation

4.103.2.1 Init() [1/2]

```
void Init (
    VoiceClient voiceClient,
    object customObj = null )
```

Initializes the [Recorder](#) component to be able to transmit audio.

Parameters

<i>voiceClient</i>	The VoiceClient to be used with this Recorder .
<i>customObj</i>	Optional user data object to be transmitted with the voice stream info

4.103.2.2 Init() [2/2]

```
void Init (
    VoiceConnection voiceConnection )
```

Initializes the [Recorder](#) component to be able to transmit audio.

Parameters

<i>voiceConnection</i>	The VoiceConnection to be used with this Recorder .
------------------------	---

4.103.2.3 ResetLocalAudio()

```
bool ResetLocalAudio ( )
```

Resets audio session and parameters locally to fix broken recording due to system configuration modifications or audio interruptions or audio routing changes.

Returns

If reset is done.

4.103.2.4 RestartRecording()

```
void RestartRecording (
    bool force = false )
```

Restarts recording if something has changed that requires this.

Parameters

<i>force</i>	Set to true if you want to restart even if this is not required (RequiresRestart = false)
--------------	---

4.103.2.5 StartRecording()

```
void StartRecording ( )
```

Starts recording.

4.103.2.6 StopRecording()

```
void StopRecording ( )
```

Stops recording.

4.103.2.7 VoiceDetectorCalibrate()

```
void VoiceDetectorCalibrate (
    int durationMs,
    Action< float > detectionEndedCallback = null )
```

Trigger voice detector calibration process. While calibrating, keep silence. [Voice](#) detector sets threshold basing on measured background noise level.

Parameters

<i>durationMs</i>	Duration of calibration in milliseconds.
<i>detectionEndedCallback</i>	Callback when VAD calibration ends.

4.103.3 Property Documentation

4.103.3.1 AudioClip

```
AudioClip AudioClip [get], [set]
```

Source audio clip.

4.103.3.2 AudioGroup

```
byte AudioGroup [get], [set]
```

Target interest group that will receive transmitted audio.

If AudioGroup != 0, recorder's audio data is sent only to clients listening to this group.

4.103.3.3 AutoStart

```
bool AutoStart [get], [set]
```

If true, automatically start recording when initialized.

4.103.3.4 Bitrate

```
int Bitrate [get], [set]
```

Outgoing audio stream bitrate.

4.103.3.5 DebugEchoMode

```
bool DebugEchoMode [get], [set]
```

If true, outgoing stream routed back to client via server same way as for remote client's streams.

4.103.3.6 Encrypt

```
bool Encrypt [get], [set]
```

If true, voice stream is sent encrypted.

4.103.3.7 FrameDuration

```
OpusCodec.FrameDuration FrameDuration [get], [set]
```

Outgoing audio stream encoder delay.

4.103.3.8 InputFactory

```
Func<IAudioDesc> InputFactory [get], [set]
```

Set the method returning new [Voice.IAudioDesc](#) instance to be assigned to a new voice created with Source set to Factory

4.103.3.9 InterestGroup

```
byte InterestGroup [get], [set]
```

Target interest group that will receive transmitted audio.

If InterestGroup != 0, recorder's audio data is sent only to clients listening to this group.

4.103.3.10 IsCurrentlyTransmitting

```
bool IsCurrentlyTransmitting [get]
```

Returns true if audio stream broadcasts.

4.103.3.11 IsInitialized

```
bool IsInitialized [get]
```

If true, this [Recorder](#) has been initialized and is ready to transmit to remote clients. Otherwise call [Init\(VoiceConnection\)](#).

4.103.3.12 IsRecording

```
bool IsRecording [get], [set]
```

Gets or sets whether this [Recorder](#) is actively recording audio to be transmitted.

4.103.3.13 LevelMeter

```
AudioUtil.ILevelMeter? LevelMeter [get]
```

Level meter utility.

4.103.3.14 LoopAudioClip

```
bool LoopAudioClip [get], [set]
```

Loop playback for audio clip sources.

4.103.3.15 MicrophoneType

```
MicType MicrophoneType [get], [set]
```

Which microphone API to use when the Source is set to Microphone.

4.103.3.16 PhotonMicrophoneDeviceId

```
int PhotonMicrophoneDeviceId [get], [set]
```

Set or get photon microphone device used for streaming.

4.103.3.17 PhotonMicrophoneEnumerator

```
AudioInEnumerator PhotonMicrophoneEnumerator [static], [get]
```

Enumerator for the available microphone devices gathered by the [Photon](#) plugin.

4.103.3.18 ReactOnSystemChanges

```
bool ReactOnSystemChanges [get], [set]
```

If true, the [Recorder](#) will automatically restart recording to recover from audio device changes.

By default, the [Recorder](#) will restart recording only when the [Recorder.SourceType](#) is `InputSourceType.Microphone` and the device being used is no longer available or valid, in some cases you may need to force restarts even if the device in use did not change. To enable this set [Recorder.SkipDeviceChangeChecks](#) to true.

4.103.3.19 RecordOnlyWhenEnabled

```
bool RecordOnlyWhenEnabled [get], [set]
```

If true, component will work only when enabled and active in hierarchy.

4.103.3.20 ReliableMode

```
bool ReliableMode [get], [set]
```

If true, stream data sent in reliable mode.

4.103.3.21 RequiresRestart

```
bool RequiresRestart [get], [protected set]
```

Returns true if something has changed in the [Recorder](#) while recording that won't take effect unless recording is restarted using [RestartRecording](#).

Think of this as a "isDirty" flag.

4.103.3.22 SamplingRate

```
SamplingRate SamplingRate [get], [set]
```

Outgoing audio stream sampling rate.

4.103.3.23 SkipDeviceChangeChecks

```
bool SkipDeviceChangeChecks [get], [set]
```

If true, restarts recording without checking if audio config/device changes affected recording.

To be used when [Recorder.ReactOnSystemChanges](#) is true.

4.103.3.24 SourceType

```
InputSourceType SourceType [get], [set]
```

Audio data source.

4.103.3.25 StopRecordingWhenPaused

```
bool StopRecordingWhenPaused [get], [set]
```

If true, stop recording when paused resume/restart when un-paused.

4.103.3.26 TransmitEnabled

```
bool TransmitEnabled [get], [set]
```

If true, audio transmission is enabled.

4.103.3.27 TrySamplingRateMatch

```
bool TrySamplingRateMatch [get], [set]
```

If true, [Recorder](#) will try to match sampling rates of microphone device and Opus encoder to avoid re sampling of audio input.

4.103.3.28 TypeConvert

```
SampleTypeConv TypeConvert [get], [set]
```

Force creation of 'short' pipeline and convert audio data to short for 'float' audio sources.

4.103.3.29 UnityMicrophoneDevice

```
string UnityMicrophoneDevice [get], [set]
```

Set or get [Unity](#) microphone device used for streaming.

4.103.3.30 UseMicrophoneTypeFallback

```
bool UseMicrophoneTypeFallback [get], [set]
```

If true, if recording fails to start with [Unity](#) microphone type, [Photon](#) microphone type is used -if available- as a fallback and vice versa.

4.103.3.31 UseOnAudioFilterRead

```
bool UseOnAudioFilterRead [get], [set]
```

If true, recording will make use of [Unity](#)'s OnAudioFiltlerRead callback from a muted local AudioSource.

If enabled, 3D sounds and voice positioning can be lost.

4.103.3.32 UserData

```
object UserData [get], [set]
```

Custom user object to be sent in the voice stream info event.

4.103.3.33 VoiceDetection

```
bool VoiceDetection [get], [set]
```

If true, voice detection enabled.

4.103.3.34 VoiceDetectionDelayMs

```
int VoiceDetectionDelayMs [get], [set]
```

Keep detected state during this time after signal level dropped below threshold. Default is 500ms

4.103.3.35 VoiceDetectionThreshold

```
float VoiceDetectionThreshold [get], [set]
```

Voice detection threshold (0..1, where 1 is full amplitude).

4.103.3.36 VoiceDetector

```
AudioUtil.IVoiceDetector? VoiceDetector [get]
```

Returns voice activity detector for recorder's audio stream.

4.103.3.37 VoiceDetectorCalibrating

```
bool VoiceDetectorCalibrating [get]
```

If true, voice detector calibration is in progress.

4.104 RemoteVoiceInfo Class Reference

Information about a remote voice (incoming stream).

Properties

- [VoiceInfo Info](#) [get]
Remote voice info.
- int [ChannelId](#) [get]
ID of channel used for transmission.
- int [PlayerId](#) [get]
Player ID of voice owner.
- byte [VoiceId](#) [get]
Voice ID (unique in the room).

4.104.1 Detailed Description

Information about a remote voice (incoming stream).

4.104.2 Property Documentation

4.104.2.1 ChannelId

```
int ChannelId [get]
```

ID of channel used for transmission.

4.104.2.2 Info

```
VoiceInfo Info [get]
```

Remote voice info.

4.104.2.3 PlayerId

```
int PlayerId [get]
```

Player ID of voice owner.

4.104.2.4 VoiceId

```
byte VoiceId [get]
```

[Voice](#) ID (unique in the room).

4.105 RemoteVoiceLink Class Reference

Public Member Functions

- **RemoteVoiceLink** ([VoiceInfo](#) info, int playerId, int voiceId, int channelId, ref [RemoteVoiceOptions](#) options)

Properties

- [VoiceInfo](#) **Info** [get]
- int **PlayerId** [get]
- int **VoiceId** [get]
- int **ChannelId** [get]

Events

- Action< [FrameOut](#)< float > > **FloatFrameDecoded**
- Action **RemoteVoiceRemoved**

4.106 RemoteVoiceOptions Struct Reference

Event Actions and other options for a remote voice (incoming stream).

Public Member Functions

- void **SetOutput** (Action< [FrameOut](#)< float >> output)
Register a method to be called when new data frame received..
- void **SetOutput** (Action< [FrameOut](#)< short >> output)
- void **SetOutput** (Action< [ImageOutputBuf](#) > output)

Properties

- Action [OnRemoteVoiceRemoveAction](#) [get, set]
Register a method to be called when the remote voice is removed.
- [IDecoder](#) **Decoder** [get, set]
Remote voice data decoder. Use to set decoder options or override it with user decoder.
- ImageFormat **OutputImageFormat** [get, set]

4.106.1 Detailed Description

Event Actions and other options for a remote voice (incoming stream).

4.106.2 Member Function Documentation

4.106.2.1 SetOutput()

```
void SetOutput (
    Action< FrameOut< float >> output )
```

Register a method to be called when new data frame received..

4.106.3 Property Documentation

4.106.3.1 Decoder

```
IDecoder Decoder [get], [set]
```

Remote voice data decoder. Use to set decoder options or override it with user decoder.

4.106.3.2 OnRemoteVoiceRemoveAction

```
Action OnRemoteVoiceRemoveAction [get], [set]
```

Register a method to be called when the remote voice is removed.

4.107 AudioUtil.Resampler< T > Class Template Reference

Sample-rate conversion Audio Processor.

Inherits [IProcessor< T >](#).

Public Member Functions

- [Resampler](#) (int dstSize, int channels)
Create a new [Resampler](#) instance.
- T[] [Process](#) (T[] buf)
Process a frame of audio data.
- void [Dispose](#) ()

Protected Attributes

- T[] [frameResampled](#)

4.107.1 Detailed Description

Sample-rate conversion Audio Processor.

This processor converts the sample-rate of the source stream. Internally, it uses `AudioUtil.Resample`.

4.107.2 Constructor & Destructor Documentation

4.107.2.1 Resampler()

```
Resampler (
    int dstSize,
    int channels )
```

Create a new `Resampler` instance.

Parameters

<i>dstSize</i>	Frame size of a destination frame. Determines output rate.
<i>channels</i>	Number of audio channels expected in both in- and output.

4.107.3 Member Function Documentation

4.107.3.1 Process()

```
T [ ] Process (
    T [ ] buf )
```

Process a frame of audio data.

Parameters

<i>buf</i>	Buffer containing input audio data
------------	------------------------------------

Returns

Buffer containing output audio data or null if frame has been discarded (VAD)

Implements `IProcessor< T >`.

4.108 SaveIncomingStreamToFile Class Reference

Inherits [VoiceComponent](#).

Protected Member Functions

- override void **Awake** ()

Additional Inherited Members

4.109 SaveOutgoingStreamToFile Class Reference

Inherits [VoiceComponent](#).

Additional Inherited Members

4.110 SavWav Class Reference

Static Public Member Functions

- static bool **Save** (string filename, AudioClip clip, bool trim=false)
- static byte[] **GetWav** (AudioClip clip, out uint length, bool trim=false)

4.111 Speaker Class Reference

Component representing remote audio stream in local scene.

Inherits [VoiceComponent](#).

Public Member Functions

- bool [StartPlayback](#) ()
Starts the audio playback of the linked incoming remote audio stream via AudioSource component.
- bool [StopPlayback](#) ()
Stops the audio playback of the linked incoming remote audio stream via AudioSource component.
- bool [RestartPlayback](#) ()
Restarts the audio playback of the linked incoming remote audio stream via AudioSource component.
- bool [SetPlaybackDelaySettings](#) ([PlaybackDelaySettings](#) pdc)
Sets the settings for the playback behaviour in case of delays.
- bool [SetPlaybackDelaySettings](#) (int low, int high, int max)
Sets the settings for the playback behaviour in case of delays.

Properties

- int **PlayDelayMs** [get, set]
- bool **IsPlaying** [get]
Is the speaker playing right now.
- int? **Lag** [get]
Smoothed difference between (jittering) stream and (clock-driven) audioOutput.
- Action< **Speaker** > **OnRemoteVoiceRemoveAction** [get, set]
Register a method to be called when remote voice removed.
- Realtime.Player **Actor** [get, set]
Per room, the connected users/players are represented with a Realtime.Player, also known as Actor.
- bool **IsLinked** [get]
*Whether or not this **Speaker** has been linked to a remote voice stream.*
- bool **PlaybackOnlyWhenEnabled** [get, set]
If true, component will work only when enabled and active in hierarchy.
- bool **PlaybackStarted** [get]
Returns if the playback is on.
- int **PlaybackDelayMinSoft** [get]
Gets the value in ms above which the audio player tries to keep the delay.
- int **PlaybackDelayMaxSoft** [get]
Gets the value in ms below which the audio player tries to keep the delay.
- int **PlaybackDelayMaxHard** [get]
Gets the value in ms that audio play delay will not exceed.

Additional Inherited Members

4.111.1 Detailed Description

Component representing remote audio stream in local scene.

4.111.2 Member Function Documentation

4.111.2.1 RestartPlayback()

```
bool RestartPlayback ( )
```

Restarts the audio playback of the linked incoming remote audio stream via AudioSource component.

Returns

True if playback is successfully restarted.

4.111.2.2 SetPlaybackDelaySettings() [1/2]

```
bool SetPlaybackDelaySettings (
    int low,
    int high,
    int max )
```

Sets the settings for the playback behaviour in case of delays.

Parameters

<i>low</i>	In milliseconds, audio player tries to keep the playback delay above this value.
<i>high</i>	In milliseconds, audio player tries to keep the playback below above this value.
<i>max</i>	In milliseconds, audio player guarantees that the playback delay never exceeds this value.

Returns

If a change has been made.

4.111.2.3 SetPlaybackDelaySettings() [2/2]

```
bool SetPlaybackDelaySettings (
    PlaybackDelaySettings pdc )
```

Sets the settings for the playback behaviour in case of delays.

Parameters

<i>pdc</i>	Playback delay configuration struct.
------------	--------------------------------------

Returns

If a change has been made.

4.111.2.4 StartPlayback()

```
bool StartPlayback ( )
```

Starts the audio playback of the linked incoming remote audio stream via AudioSource component.

Returns

True if playback is successfully started.

4.111.2.5 StopPlayback()

```
bool StopPlayback ( )
```

Stops the audio playback of the linked incoming remote audio stream via AudioSource component.

Returns

True if playback is successfully stopped.

4.111.3 Property Documentation

4.111.3.1 Actor

```
Realtime.Player Actor [get], [set]
```

Per room, the connected users/players are represented with a `Realtime.Player`, also known as Actor.

[Photon Voice](#) calls this Actor, to avoid a name-clash with the Player class in [Voice](#).

4.111.3.2 IsLinked

```
bool IsLinked [get]
```

Whether or not this [Speaker](#) has been linked to a remote voice stream.

4.111.3.3 IsPlaying

```
bool IsPlaying [get]
```

Is the speaker playing right now.

4.111.3.4 Lag

```
int? Lag [get]
```

Smoothed difference between (jittering) stream and (clock-driven) audioOutput.

4.111.3.5 OnRemoteVoiceRemoveAction

```
Action<Speaker> OnRemoteVoiceRemoveAction [get], [set]
```

Register a method to be called when remote voice removed.

4.111.3.6 PlaybackDelayMaxHard

```
int PlaybackDelayMaxHard [get]
```

Gets the value in ms that audio play delay will not exceed.

4.111.3.7 PlaybackDelayMaxSoft

```
int PlaybackDelayMaxSoft [get]
```

Gets the value in ms below which the audio player tries to keep the delay.

4.111.3.8 PlaybackDelayMinSoft

```
int PlaybackDelayMinSoft [get]
```

Gets the value in ms above which the audio player tries to keep the delay.

4.111.3.9 PlaybackOnlyWhenEnabled

```
bool PlaybackOnlyWhenEnabled [get], [set]
```

If true, component will work only when enabled and active in hierarchy.

4.111.3.10 PlaybackStarted

```
bool PlaybackStarted [get]
```

Returns if the playback is on.

4.112 TestTone Class Reference

Inherits MonoBehaviour.

4.113 AudioUtil.ToneAudioPusher< T > Class Template Reference

[IAudioPusher](#) that provides a constant tone signal.

Inherits [IAudioPusher< T >](#).

Public Member Functions

- [ToneAudioPusher](#) (int frequency=440, int bufSizeMs=100, int samplingRate=48000, int channels=2)
Create a new [ToneAudioReader](#) instance
- void [SetCallback](#) (Action< T[]> callback, [ObjectFactory](#)< T[], int > bufferFactory)
Set the callback function used for pushing data
- void **Dispose** ()

Properties

- int **Channels** [get]
- int **SamplingRate** [get]
- string **Error** [get]

4.113.1 Detailed Description

[IAudioPusher](#) that provides a constant tone signal.

4.113.2 Constructor & Destructor Documentation

4.113.2.1 ToneAudioPusher()

```
ToneAudioPusher (
    int frequency = 440,
    int bufSizeMs = 100,
    int samplingRate = 48000,
    int channels = 2 )
```

Create a new [ToneAudioReader](#) instance

Parameters

<i>frequency</i>	Frequency of the generated tone (in Hz).
<i>bufSizeMs</i>	Size of buffers to push (in milliseconds).
<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>channels</i>	Number of channels in the audio signal.

4.113.3 Member Function Documentation

4.113.3.1 SetCallback()

```
void SetCallback (
    Action< T[] > callback,
    ObjectFactory< T[], int > bufferFactory )
```

Set the callback function used for pushing data

Parameters

<i>callback</i>	Callback function to use
<i>localVoice</i>	Outgoing audio stream, for context

Implements [IAudioPusher< T >](#).

4.114 ToneAudioReader Class Reference

Inherits [IAudioReader< float >](#).

Public Member Functions

- void **Dispose** ()
- bool **Read** (float[] buf)

Properties

- int **Channels** [get]
- int **SamplingRate** [get]
- string **Error** [get]

4.115 AudioUtil.ToneAudioReader< T > Class Template Reference

[IAudioReader](#) that provides a constant tone signal.

Inherits [IAudioReader< T >](#).

Public Member Functions

- [ToneAudioReader](#) (Func< double > clockSec=null, double frequency=440, int samplingRate=48000, int channels=2)

Create a new [ToneAudioReader](#) instance

- void **Dispose** ()
- bool **Read** (T[] buf)

Fill full given frame buffer with source uncompressed data or return false if not enough such data.

Properties

- int [Channels](#) [get]
Number of channels in the audio signal.
- int [SamplingRate](#) [get]
Sampling rate of the audio signal (in Hz).
- string [Error](#) [get]
If not null, audio object is in invalid state.

4.115.1 Detailed Description

[IAudioReader](#) that provides a constant tone signal.

See also [MicWrapper](#) and [AudioClipWrapper](#). Because of current resampling algorithm, the tone is distorted if [SamplingRate](#) does not equal encoder sampling rate.

4.115.2 Constructor & Destructor Documentation

4.115.2.1 ToneAudioReader()

```
ToneAudioReader (
    Func< double > clockSec = null,
    double frequency = 440,
    int samplingRate = 48000,
    int channels = 2 )
```

Create a new [ToneAudioReader](#) instance

Parameters

<i>clockSec</i>	Function to get current time in seconds. In Unity , pass in '()' => AudioSettings.dspTime for better results.
<i>frequency</i>	Frequency of the generated tone (in Hz).
<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>channels</i>	Number of channels in the audio signal.

4.115.3 Member Function Documentation

4.115.3.1 Read()

```
bool Read (
    T[] buffer )
```

Fill full given frame buffer with source uncompressed data or return false if not enough such data.

Parameters

<i>buffer</i>	Buffer to fill.
---------------	-----------------

Returns

True if buffer was filled successfully, false otherwise.

Implements [IDataReader< T >](#).

4.115.4 Property Documentation

4.115.4.1 Channels

```
int Channels [get]
```

Number of channels in the audio signal.

4.115.4.2 Error

```
string Error [get]
```

If not null, audio object is in invalid state.

4.115.4.3 SamplingRate

```
int SamplingRate [get]
```

Sampling rate of the audio signal (in Hz).

4.116 UnityAndroidAudioInAEC Class Reference

Inherits [IAudioPusher< short >](#), and [IResettable](#).

Public Member Functions

- **UnityAndroidAudioInAEC** ([Voice.ILogger](#) logger, bool enableAEC=false, bool enableAGC=false, bool enableNS=false)
- void **SetCallback** (Action< short[]> callback, [ObjectFactory](#)< short[], int > bufferFactory)
- void **Reset** ()
- void **Dispose** ()

Properties

- int **Channels** [get]
- int **SamplingRate** [get]
- string **Error** [get]

4.117 UnityAudioOut Class Reference

Inherits [IAudioOut< float >](#).

Classes

- struct [PlayDelayConfig](#)

Public Member Functions

- **UnityAudioOut** (AudioSource audioSource, [PlayDelayConfig](#) playDelayConfig, [ILogger](#) logger, string logPrefix, bool debugInfo)
- void **Start** (int frequency, int channels, int frameSamples)
- void **Service** ()
- void **Push** (float[] frame)
- void **Flush** ()
- void **Stop** ()

Static Public Attributes

- const int **FRAME_POOL_CAPACITY** = 50

Properties

- int? **Lag** [get]
- bool **IsPlaying** [get]

4.118 UnityMicrophone Class Reference

A wrapper around UnityEngine.Microphone to be able to safely use Microphone and compile for WebGL.

Static Public Member Functions

- static void **End** (string deviceName)
- static void **GetDeviceCaps** (string deviceName, out int minFreq, out int maxFreq)
- static int **GetPosition** (string deviceName)
- static bool **IsRecording** (string deviceName)
- static AudioClip **Start** (string deviceName, bool loop, int lengthSec, int frequency)

Properties

- static string[] **devices** [get]

4.118.1 Detailed Description

A wrapper around UnityEngine.Microphone to be able to safely use Microphone and compile for WebGL.

4.119 UnsupportedCodecException Class Reference

Exception thrown if an unsupported codec is encountered.

Inherits Exception.

Public Member Functions

- [UnsupportedCodecException](#) (string info, [Codec](#) codec, [ILogger](#) logger)
Create a new [UnsupportedCodecException](#).

4.119.1 Detailed Description

Exception thrown if an unsupported codec is encountered.

PhotonVoice currently only supports one Codec, [Codec.AudioOpus](#).

4.119.2 Constructor & Destructor Documentation

4.119.2.1 UnsupportedCodecException()

```
UnsupportedCodecException (
    string info,
    Codec codec,
    ILogger logger )
```

Create a new [UnsupportedCodecException](#).

Parameters

<i>info</i>	The info prepending standard message.
<i>codec</i>	The codec actually encountered.
<i>logger</i>	Loogger.

4.120 UnsupportedSampleTypeException Class Reference

Exception thrown if an unsupported audio sample type is encountered.

Inherits Exception.

Public Member Functions

- [UnsupportedSampleTypeException](#) (Type t)
Create a new [UnsupportedSampleTypeException](#).

4.120.1 Detailed Description

Exception thrown if an unsupported audio sample type is encountered.

PhotonVoice generally supports 32-bit floating point ("float") or 16-bit signed integer ("short") audio, but it usually won't be converted automatically due to the high CPU overhead (and potential loss of precision) involved.

4.120.2 Constructor & Destructor Documentation

4.120.2.1 UnsupportedSampleTypeException()

```
UnsupportedSampleTypeException (
    Type t )
```

Create a new [UnsupportedSampleTypeException](#).

Parameters

<i>t</i>	The sample type actually encountered.
----------	---------------------------------------

4.121 OpusCodec.Util Class Reference

4.122 VoiceClient Class Reference

[Voice](#) client interact with other clients on network via [IVoiceTransport](#).

Inherits IDisposable.

Public Member Functions

- delegate void [RemoteVoiceInfoDelegate](#) (int channelId, int playerId, byte voiceId, [VoiceInfo](#) voiceInfo, ref [RemoteVoiceOptions](#) options)
Remote voice info event delegate.
- IEnumerable< [LocalVoice](#) > [LocalVoicesInChannel](#) (int channelId)
Iterates through copy of all local voices list of given channel.
- void **LogRemoteVoiceStates** ()
- void [Service](#) ()
This method dispatches all available incoming commands and then sends this client's outgoing commands. Call this method regularly (2..20 times a second).
- [LocalVoice](#) [CreateLocalVoice](#) ([VoiceInfo](#) voiceInfo, int channelId=0, [IEncoder](#) encoder=null)
Creates basic outgoing stream w/o data processing support. Provided encoder should generate output data stream.
- [LocalVoiceFramed](#)< T > [CreateLocalVoiceFramed](#)< T > ([VoiceInfo](#) voiceInfo, int frameSize, int channelId=0, [IEncoder](#) encoder=null)
Creates outgoing stream consuming sequence of values passed in array buffers of arbitrary length which repacked in frames of constant length for further processing and encoding.
- [LocalVoiceAudio](#)< T > [CreateLocalVoiceAudio](#)< T > ([VoiceInfo](#) voiceInfo, [IAudioDesc](#) audioSourceDesc, [IEncoder](#) encoder, int channelId=0)
Creates outgoing audio stream. Adds audio specific features (e.g. resampling, level meter) to processing pipeline and to returning stream handler.
- [LocalVoice](#) [CreateLocalVoiceAudioFromSource](#) ([VoiceInfo](#) voiceInfo, [IAudioDesc](#) source, [AudioSampleType](#) sampleType, [IEncoder](#) encoder=null, int channelId=0)
Creates outgoing audio stream of type automatically assigned and adds procedures (callback or serviceable) for consuming given audio source data. Adds audio specific features (e.g. resampling, level meter) to processing pipeline and to returning stream handler.
- void [RemoveLocalVoice](#) ([LocalVoice](#) voice)
Removes local voice (outgoing data stream).

Parameters

voice	Handler of outgoing stream to be removed.
-------	---

- void **Dispose** ()

Properties

- int [FramesLost](#) [get, set]
Lost frames counter.
- int [FramesReceived](#) [get]
Received frames counter.
- int [FramesSent](#) [get]
Sent frames counter.
- int [FramesSentBytes](#) [get]
Sent frames bytes counter.
- int [RoundTripTime](#) [get]
Average time required voice packet to return to sender.
- int [RoundTripTimeVariance](#) [get]
Average round trip time variation.
- bool [SuppressInfoDuplicateWarning](#) [get, set]
Do not log warning when duplicate info received.
- [RemoteVoiceInfoDelegate](#) [OnRemoteVoiceInfoAction](#) [get, set]

Register a method to be called when remote voice info arrived (after join or new new remote voice creation). Metod parameters: (int channelId, int playerId, byte voiceld, [VoiceInfo](#) voiceInfo, ref [RemoteVoiceOptions](#) options);

- int [DebugLostPercent](#) [get, set]
Lost frames simulation ratio.
- IEnumerable< [LocalVoice](#) > [LocalVoices](#) [get]
Iterates through copy of all local voices list.
- IEnumerable< [RemoteVoiceInfo](#) > [RemoteVoiceInfos](#) [get]
Iterates through all remote voices infos.

4.122.1 Detailed Description

[Voice](#) client interact with other clients on network via [IVoiceTransport](#).

4.122.2 Member Function Documentation

4.122.2.1 CreateLocalVoice()

```
LocalVoice CreateLocalVoice (
    VoiceInfo voiceInfo,
    int channelId = 0,
    IEncoder encoder = null )
```

Creates basic outgoing stream w/o data processing support. Provided encoder should generate output data stream.

Parameters

<i>voiceInfo</i>	Outgoing stream parameters. Set applicable fields to read them by encoder and by receiving client when voice created.
<i>channelId</i>	Transport channel specific to transport.
<i>encoder</i>	Encoder producing the stream.

Returns

Outgoing stream handler.

4.122.2.2 CreateLocalVoiceAudio< T >()

```
LocalVoiceAudio<T> CreateLocalVoiceAudio< T > (
    VoiceInfo voiceInfo,
```

```
IAudioDesc audioSourceDesc,
IEncoder encoder,
int channelId = 0 )
```

Creates outgoing audio stream. Adds audio specific features (e.g. resampling, level meter) to processing pipeline and to returning stream handler.

Template Parameters

<i>T</i>	Element type of audio array buffers.
----------	--------------------------------------

Parameters

<i>voiceInfo</i>	Outgoing audio stream parameters. Set applicable fields to read them by encoder and by receiving client when voice created.
<i>channelId</i>	Transport channel specific to transport.
<i>encoder</i>	Audio encoder. Set to null to use default Opus encoder.

Returns

Outgoing stream handler.

audioSourceDesc.SamplingRate and voiceInfo.SamplingRate may do not match. Automatic resampling will occur in this case.

4.122.2.3 CreateLocalVoiceAudioFromSource()

```
LocalVoice CreateLocalVoiceAudioFromSource (
    VoiceInfo voiceInfo,
    IAudioDesc source,
    AudioSampleType sampleType,
    IEncoder encoder = null,
    int channelId = 0 )
```

Creates outgoing audio stream of type automatically assigned and adds procedures (callback or serviceable) for consuming given audio source data. Adds audio specific features (e.g. resampling, level meter) to processing pipeline and to returning stream handler.

Parameters

<i>voiceInfo</i>	Outgoing audio stream parameters. Set applicable fields to read them by encoder and by receiving client when voice created.
<i>source</i>	Streaming audio source.
<i>sampleType</i>	Voice's audio sample type. If does not match source audio sample type, conversion will occur.
<i>channelId</i>	Transport channel specific to transport.
<i>encoder</i>	Audio encoder. Set to null to use default Opus encoder.

Returns

Outgoing stream handler.

audioSourceDesc.SamplingRate and voiceInfo.SamplingRate may do not match. Automatic resampling will occur in this case.

4.122.2.4 CreateLocalVoiceFramed< T >()

```
LocalVoiceFramed<T> CreateLocalVoiceFramed< T > (
    VoiceInfo voiceInfo,
    int frameSize,
    int channelId = 0,
    IEncoder encoder = null )
```

Creates outgoing stream consuming sequence of values passed in array buffers of arbitrary length which repacked in frames of constant length for further processing and encoding.

Template Parameters

<i>T</i>	Type of data consumed by outgoing stream (element type of array buffers).
----------	---

Parameters

<i>voiceInfo</i>	Outgoing stream parameters. Set applicable fields to read them by encoder and by receiving client when voice created.
<i>frameSize</i>	Size of buffer LocalVoiceFramed repacks input data stream to.
<i>channelId</i>	Transport channel specific to transport.
<i>encoder</i>	Encoder compressing data stream in pipeline.

Returns

Outgoing stream handler.

4.122.2.5 LocalVoicesInChannel()

```
IEnumerable<LocalVoice> LocalVoicesInChannel (
    int channelId )
```

Iterates through copy of all local voices list of given channel.

4.122.2.6 RemoteVoiceInfoDelegate()

```
delegate void RemoteVoiceInfoDelegate (
    int channelId,
    int playerId,
    byte voiceId,
    VoiceInfo voiceInfo,
    ref RemoteVoiceOptions options )
```

Remote voice info event delegate.

4.122.2.7 RemoveLocalVoice()

```
void RemoveLocalVoice (
    LocalVoice voice )
```

Removes local voice (outgoing data stream).

Parameters

<i>voice</i>	Handler of outgoing stream to be removed.
--------------	---

4.122.2.8 Service()

```
void Service ( )
```

This method dispatches all available incoming commands and then sends this client's outgoing commands. Call this method regularly (2..20 times a second).

4.122.3 Property Documentation

4.122.3.1 DebugLostPercent

```
int DebugLostPercent [get], [set]
```

Lost frames simulation ratio.

4.122.3.2 FramesLost

```
int FramesLost [get], [set]
```

Lost frames counter.

4.122.3.3 FramesReceived

```
int FramesReceived [get]
```

Received frames counter.

4.122.3.4 FramesSent

```
int FramesSent [get]
```

Sent frames counter.

4.122.3.5 FramesSentBytes

```
int FramesSentBytes [get]
```

Sent frames bytes counter.

4.122.3.6 LocalVoices

```
IEnumerable<LocalVoice> LocalVoices [get]
```

Iterates through copy of all local voices list.

4.122.3.7 OnRemoteVoiceInfoAction

```
RemoteVoiceInfoDelegate OnRemoteVoiceInfoAction [get], [set]
```

Register a method to be called when remote voice info arrived (after join or new new remote voice creation). Metod parameters: (int channelId, int playerId, byte voiceId, [VoiceInfo](#) voiceInfo, ref [RemoteVoiceOptions](#) options);

4.122.3.8 RemoteVoiceInfos

```
IEnumerable<RemoteVoiceInfo> RemoteVoiceInfos [get]
```

Iterates through all remote voices infos.

4.122.3.9 RoundTripTime

```
int RoundTripTime [get]
```

Average time required voice packet to return to sender.

4.122.3.10 RoundTripTimeVariance

```
int RoundTripTimeVariance [get]
```

Average round trip time variation.

4.122.3.11 SuppressInfoDuplicateWarning

```
bool SuppressInfoDuplicateWarning [get], [set]
```

Do not log warning when duplicate info received.

4.123 VoiceComponent Class Reference

Inherits MonoBehaviour, and [ILoggableDependent](#).

Inherited by [PhotonVoiceView](#), [Recorder](#), [Speaker](#), [MicAmplifier](#), [SaveIncomingStreamToFile](#), [SaveOutgoingStreamToFile](#), and [WebRtcAudioDsp](#).

Protected Member Functions

- virtual void **Awake** ()

Protected Attributes

- DebugLevel **logLevel** = DebugLevel.INFO

Properties

- [VoiceLogger](#) **Logger** [get, protected set]
- DebugLevel **LogLevel** [get, set]
- bool **IgnoreGlobalLogLevel** [get, set]

4.124 VoiceConnection Class Reference

Component that represents a client voice connection to [Photon](#) Servers.

Inherits [ConnectionHandler](#), and [ILoggable](#).

Inherited by [PhotonVoiceNetwork](#).

Public Member Functions

- bool [ConnectUsingSettings](#) (AppSettings overwriteSettings=null)
Connect to [Photon](#) server using [Settings](#)
- void [InitRecorder](#) ([Recorder](#) rec)
Initializes the [Recorder](#) component to be able to transmit audio.
- void [SetPlaybackDelaySettings](#) ([PlaybackDelaySettings](#) gpds)
Sets the global configuration for the playback behaviour in case of delays.
- void [SetGlobalPlaybackDelaySettings](#) (int low, int high, int max)
Sets the global configuration for the playback behaviour in case of delays.

Public Attributes

- AppSettings [Settings](#)
Settings to be used by this voice connection
- Func< int, byte, object, [Speaker](#) > [SpeakerFactory](#)
Special factory to link [Speaker](#) components with incoming remote audio streams
- float [MinimalTimeScaleToDispatchInFixedUpdate](#) = -1f
Configures the minimal Time.timeScale at which [Voice](#) client will dispatch incoming messages within LateUpdate.
- bool [AutoCreateSpeakerIfNotFound](#) = true
Auto instantiate a GameObject and attach a [Speaker](#) component to link to a remote audio stream if no candidate could be found

Protected Member Functions

- override void **Awake** ()
- virtual void **Update** ()
- virtual void **FixedUpdate** ()
- void [Dispatch](#) ()
Dispatches incoming network messages for [Voice](#) client. Called in FixedUpdate or LateUpdate.
- override void **OnDisable** ()
- virtual void **OnDestroy** ()
- virtual [Speaker](#) **SimpleSpeakerFactory** (int playerId, byte voiceId, object userData)
- virtual void **OnVoiceStateChanged** ([ClientState](#) fromState, [ClientState](#) toState)
- void **CalcStatistics** ()
- void **LinkSpeaker** ([Speaker](#) speaker, [RemoteVoiceLink](#) remoteVoice)

Protected Attributes

- List< [RemoteVoiceLink](#) > **cachedRemoteVoices** = new List<[RemoteVoiceLink](#)>()

Properties

- [VoiceLogger](#) **Logger** [get, protected set]
Logger used by this component
- DebugLevel [LogLevel](#) [get, set]
Log level for this component
- new [LoadBalancingTransport](#) **Client** [get]
- [VoiceClient](#) **VoiceClient** [get]
Returns underlying [Photon Voice](#) client.
- ClientState [ClientState](#) [get]
Returns [Photon Voice](#) client state.
- float [FramesReceivedPerSecond](#) [get]
Number of frames received per second.
- float [FramesLostPerSecond](#) [get]
Number of frames lost per second.
- float [FramesLostPercent](#) [get]
Percentage of lost frames.
- GameObject [SpeakerPrefab](#) [get, set]
Prefab that contains [Speaker](#) component to be instantiated when receiving a new remote audio source info
- [Recorder](#) **PrimaryRecorder** [get, set]
Main [Recorder](#) to be used for transmission by default
- DebugLevel **GlobalRecordersLogLevel** [get, set]
- DebugLevel **GlobalSpeakersLogLevel** [get, set]
- int **GlobalPlaybackDelay** [get, set]
- string [BestRegionSummaryInPreferences](#) [get, set]
Used to store and access the "Best Region Summary" in the Player Preferences.
- int [GlobalPlaybackDelayMinSoft](#) [get]
Gets the global value in ms above which the audio player tries to keep the delay.
- int [GlobalPlaybackDelayMaxSoft](#) [get]
Gets the global value in ms below which the audio player tries to keep the delay.
- int [GlobalPlaybackDelayMaxHard](#) [get]
Gets the global value in ms that audio play delay will not exceed.

Events

- Action< [Speaker](#) > **SpeakerLinked**
Fires when a speaker has been linked to a remote audio stream
- Action< [RemoteVoiceLink](#) > **RemoteVoiceAdded**
Fires when a remote voice stream is added

4.124.1 Detailed Description

Component that represents a client voice connection to [Photon](#) Servers.

4.124.2 Member Function Documentation

4.124.2.1 ConnectUsingSettings()

```
bool ConnectUsingSettings (
    AppSettings overwriteSettings = null )
```

Connect to [Photon](#) server using [Settings](#)

Parameters

<i>overwriteSettings</i>	Overwrites Settings before connecting
--------------------------	---

Returns

If true voice connection command was sent from client

4.124.2.2 Dispatch()

```
void Dispatch ( ) [protected]
```

Dispatches incoming network messages for [Voice](#) client. Called in FixedUpdate or LateUpdate.

It may make sense to dispatch incoming messages, even if the timeScale is near 0. That can be configured with [MinimalTimeScaleToDispatchInFixedUpdate](#).

Without dispatching messages, [Voice](#) client won't change state and does not handle updates.

4.124.2.3 InitRecorder()

```
void InitRecorder (
    Recorder rec )
```

Initializes the [Recorder](#) component to be able to transmit audio.

Parameters

<i>rec</i>	The Recorder to be initialized.
------------	---

4.124.2.4 SetGlobalPlaybackDelaySettings()

```
void SetGlobalPlaybackDelaySettings (
```

```
int low,
int high,
int max )
```

Sets the global configuration for the playback behaviour in case of delays.

Parameters

<i>low</i>	In milliseconds, audio player tries to keep the playback delay above this value.
<i>high</i>	In milliseconds, audio player tries to keep the playback below above this value.
<i>max</i>	In milliseconds, audio player guarantees that the playback delay never exceeds this value.

4.124.2.5 SetPlaybackDelaySettings()

```
void SetPlaybackDelaySettings (
    PlaybackDelaySettings gpds )
```

Sets the global configuration for the playback behaviour in case of delays.

Parameters

<i>gpds</i>	Playback delay configuration struct.
-------------	--------------------------------------

4.124.3 Member Data Documentation

4.124.3.1 AutoCreateSpeakerIfNotFound

```
bool AutoCreateSpeakerIfNotFound = true
```

Auto instantiate a `GameObject` and attach a [Speaker](#) component to link to a remote audio stream if no candidate could be found

4.124.3.2 MinimalTimeScaleToDispatchInFixedUpdate

```
float MinimalTimeScaleToDispatchInFixedUpdate = -1f
```

Configures the minimal `Time.timeScale` at which [Voice](#) client will dispatch incoming messages within `LateUpdate`.

It may make sense to dispatch incoming messages, even if the `timeScale` is near 0. In some cases, stopping the game time makes sense, so this option defaults to -1f, which is "off". Without dispatching messages, [Voice](#) client won't change state and does not handle updates.

4.124.3.3 Settings

`AppSettings Settings`

Settings to be used by this voice connection

4.124.3.4 SpeakerFactory

`Func<int, byte, object, Speaker> SpeakerFactory`

Special factory to link [Speaker](#) components with incoming remote audio streams

4.124.4 Property Documentation

4.124.4.1 BestRegionSummaryInPreferences

`string BestRegionSummaryInPreferences [get], [set]`

Used to store and access the "Best Region Summary" in the Player Preferences.

4.124.4.2 ClientState

`ClientState ClientState [get]`

Returns [Photon Voice](#) client state.

4.124.4.3 FramesLostPercent

`float FramesLostPercent [get]`

Percentage of lost frames.

4.124.4.4 FramesLostPerSecond

`float FramesLostPerSecond [get]`

Number of frames lost per second.

4.124.4.5 FramesReceivedPerSecond

```
float FramesReceivedPerSecond [get]
```

Number of frames received per second.

4.124.4.6 GlobalPlaybackDelayMaxHard

```
int GlobalPlaybackDelayMaxHard [get]
```

Gets the global value in ms that audio play delay will not exceed.

4.124.4.7 GlobalPlaybackDelayMaxSoft

```
int GlobalPlaybackDelayMaxSoft [get]
```

Gets the global value in ms below which the audio player tries to keep the delay.

4.124.4.8 GlobalPlaybackDelayMinSoft

```
int GlobalPlaybackDelayMinSoft [get]
```

Gets the global value in ms above which the audio player tries to keep the delay.

4.124.4.9 Logger

```
VoiceLogger Logger [get], [protected set]
```

[Logger](#) used by this component

4.124.4.10 LogLevel

```
DebugLevel LogLevel [get], [set]
```

Log level for this component

4.124.4.11 PrimaryRecorder

[Recorder](#) PrimaryRecorder [get], [set]

Main [Recorder](#) to be used for transmission by default

4.124.4.12 SpeakerPrefab

GameObject SpeakerPrefab [get], [set]

Prefab that contains [Speaker](#) component to be instantiated when receiving a new remote audio source info

4.124.4.13 VoiceClient

[VoiceClient](#) VoiceClient [get]

Returns underlying [Photon Voice](#) client.

4.124.5 Event Documentation

4.124.5.1 RemoteVoiceAdded

Action<[RemoteVoiceLink](#)> RemoteVoiceAdded

Fires when a remote voice stream is added

4.124.5.2 SpeakerLinked

Action<[Speaker](#)> SpeakerLinked

Fires when a speaker has been linked to a remote audio stream

4.125 AudioUtil.VoiceDetector< T > Class Template Reference

Simple voice activity detector triggered by signal level.

Inherits [IProcessor< T >](#), and [AudioUtil.IVoiceDetector](#).

Public Member Functions

- abstract T[] [Process](#) (T[] buf)
Process a frame of audio data.
- void **Dispose** ()

Protected Attributes

- float **norm**
- float **threshold**
- int **activityDelay**
- int **autoSilenceCounter** = 0
- int **valuesCountPerSec**
- int **activityDelayValuesCount**

Properties

- bool [On](#) [get, set]
If true, voice detection enabled.
- float [Threshold](#) [get, set]
Voice detected as soon as signal level exceeds threshold.
- bool [Detected](#) [get, protected set]
If true, voice detected.
- DateTime [DetectedTime](#) [get]
Last time when switched to detected state.
- int [ActivityDelayMs](#) [get, set]
Keep detected state during this time after signal level dropped below threshold.

Events

- Action [OnDetected](#)
Called when switched to detected state.

4.125.1 Detailed Description

Simple voice activity detector triggered by signal level.

4.125.2 Member Function Documentation

4.125.2.1 Process()

```
abstract T [] Process (  
    T[] buf ) [pure virtual]
```

Process a frame of audio data.

Parameters

<i>buf</i>	Buffer containing input audio data
------------	------------------------------------

Returns

Buffer containing output audio data or null if frame has been discarded (VAD)

Implements [IProcessor< T >](#).

4.125.3 Property Documentation

4.125.3.1 ActivityDelayMs

```
int ActivityDelayMs [get], [set]
```

Keep detected state during this time after signal level dropped below threshold.

4.125.3.2 Detected

```
bool Detected [get], [protected set]
```

If true, voice detected.

4.125.3.3 DetectedTime

```
DateTime DetectedTime [get]
```

Last time when switched to detected state.

4.125.3.4 On

```
bool On [get], [set]
```

If true, voice detection enabled.

4.125.3.5 Threshold

float Threshold [get], [set]

Voice detected as soon as signal level exceeds threshold.

4.125.4 Event Documentation

4.125.4.1 OnDetected

Action OnDetected

Called when switched to detected state.

4.126 AudioUtil.VoiceDetectorCalibration< T > Class Template Reference

Calibration Utility for Voice Detector

Inherits [IProcessor< T >](#).

Public Member Functions

- [VoiceDetectorCalibration](#) ([IVoiceDetector](#) voiceDetector, [ILevelMeter](#) levelMeter, int samplingRate, int channels)
Create new [VoiceDetectorCalibration](#) instance.
- void [Calibrate](#) (int durationMs, Action< float > onCalibrated=null)
Start calibration.
- T[] [Process](#) (T[] buf)
Process a frame of audio data.
- void **Dispose** ()

Protected Attributes

- int **calibrateCount**

Properties

- bool **IsCalibrating** [get]

4.126.1 Detailed Description

Calibration Utility for [Voice](#) Detector

. Using this audio processor, you can calibrate the [IVoiceDetector.Threshold](#).

4.126.2 Constructor & Destructor Documentation

4.126.2.1 VoiceDetectorCalibration()

```
VoiceDetectorCalibration (
    IVoiceDetector voiceDetector,
    ILevelMeter levelMeter,
    int samplingRate,
    int channels )
```

Create new [VoiceDetectorCalibration](#) instance.

Parameters

<i>voiceDetector</i>	Voice Detector to calibrate.
<i>levelMeter</i>	Level Meter to look at for calibration.
<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>numChannels</i>	Number of channels in the audio signal.

4.126.3 Member Function Documentation

4.126.3.1 Calibrate()

```
void Calibrate (
    int durationMs,
    Action< float > onCalibrated = null )
```

Start calibration.

Parameters

<i>durationMs</i>	Duration of the calibration procedure (in milliseconds).
-------------------	--

This activates the Calibration process. It will reset the given [LevelMeter](#)'s AccumAvgPeakAmp (accumulated average peak amplitude), and when the duration has passed, use it for the [VoiceDetector](#)'s detection threshold.

4.126.3.2 Process()

```
T [] Process (
    T[] buf )
```

Process a frame of audio data.

Parameters

<i>buf</i>	Buffer containing input audio data
------------	------------------------------------

Returns

Buffer containing output audio data or null if frame has been discarded (VAD)

Implements [IProcessor< T >](#).

4.127 AudioUtil.VoiceDetectorDummy Class Reference

Dummy [VoiceDetector](#) that doesn't actually do anything.

Inherits [AudioUtil.IVoiceDetector](#).

Properties

- bool **On** [get, set]
- float **Threshold** [get, set]
- bool **Detected** [get]
- int **ActivityDelayMs** [get, set]
- DateTime **DetectedTime** [get]
- Action **OnDetected**

Additional Inherited Members

4.127.1 Detailed Description

Dummy [VoiceDetector](#) that doesn't actually do anything.

4.128 AudioUtil.VoiceDetectorFloat Class Reference

[VoiceDetector](#) specialization for float audio.

Inherits [AudioUtil.VoiceDetector< float >](#).

Public Member Functions

- [VoiceDetectorFloat](#) (int `samplingRate`, int `numChannels`)
Create a new [VoiceDetectorFloat](#) instance.
- override float[] **Process** (float[] `buffer`)

Additional Inherited Members

4.128.1 Detailed Description

[VoiceDetector](#) specialization for float audio.

4.128.2 Constructor & Destructor Documentation

4.128.2.1 VoiceDetectorFloat()

```
VoiceDetectorFloat (
    int samplingRate,
    int numChannels )
```

Create a new [VoiceDetectorFloat](#) instance.

Parameters

<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>numChannels</i>	Number of channels in the audio signal.

4.129 AudioUtil.VoiceDetectorShort Class Reference

[VoiceDetector](#) specialization for float audio.

Inherits [AudioUtil.VoiceDetector< short >](#).

Public Member Functions

- [VoiceDetectorShort](#) (int `samplingRate`, int `numChannels`)
Create a new [VoiceDetectorFloat](#) instance
- override short[] **Process** (short[] `buffer`)

Additional Inherited Members

4.129.1 Detailed Description

[VoiceDetector](#) specialization for float audio.

4.129.2 Constructor & Destructor Documentation

4.129.2.1 VoiceDetectorShort()

```
VoiceDetectorShort (
    int samplingRate,
    int numChannels )
```

Create a new [VoiceDetectorFloat](#) instance

Parameters

<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>numChannels</i>	Number of channels in the audio signal.

4.130 VoiceEvent Class Reference

Static Public Attributes

- const byte [Code](#) = 202
Single event used for voice communications.
- const byte **FrameCode** = 203

4.130.1 Member Data Documentation

4.130.1.1 Code

```
const byte Code = 202 [static]
```

Single event used for voice communications.

Change if it conflicts with other event codes used in the same [Photon](#) room.

4.131 VoiceInfo Struct Reference

Describes stream properties.

Public Member Functions

- override string **ToString** ()

Static Public Member Functions

- static [VoiceInfo CreateAudioOpus](#) (POpusCodec.Enums.SamplingRate samplingRate, int channels, Opus↔Codec.FrameDuration frameDurationUs, int bitrate, object userdata=null)
Create stream info for an Opus audio stream.
- static [VoiceInfo CreateAudio](#) (Codec codec, int samplingRate, int channels, int frameDurationUs, object userdata=null)
Create stream info for an Opus audio stream.

Properties

- [Codec Codec](#) [get, set]
- int [SamplingRate](#) [get, set]
Audio sampling rate (frequency, in Hz).
- int [Channels](#) [get, set]
Number of channels.
- int [FrameDurationUs](#) [get, set]
Uncompressed frame (audio packet) size in microseconds.
- int [Bitrate](#) [get, set]
Target bitrate (in bits/second).
- int [Width](#) [get, set]
Video width.
- int [Height](#) [get, set]
Video height
- int [FPS](#) [get, set]
Video frames per second
- int [KeyFrameInt](#) [get, set]
Video keyframe interval in frames
- object [UserData](#) [get, set]
Optional user data. Should be serializable by [Photon](#).
- int [FrameDurationSamples](#) [get]
Uncompressed frame (data packet) size in samples.
- int [FrameSize](#) [get]
Uncompressed frame (data packet) array size.

4.131.1 Detailed Description

Describes stream properties.

4.131.2 Member Function Documentation

4.131.2.1 CreateAudio()

```
static VoiceInfo CreateAudio (
    Codec codec,
    int samplingRate,
    int channels,
    int frameDurationUs,
    object userdata = null ) [static]
```

Create stream info for an Opus audio stream.

Parameters

<i>samplingRate</i>	Audio sampling rate.
<i>channels</i>	Number of channels.
<i>frameDurationUs</i>	Uncompressed frame (audio packet) size in microseconds.
<i>bitrate</i>	Stream bitrate (in bits/second).
<i>userdata</i>	Optional user data. Should be serializable by Photon .

Returns

[VoiceInfo](#) instance.

4.131.2.2 CreateAudioOpus()

```
static VoiceInfo CreateAudioOpus (  
    POpusCodec.Enums.SamplingRate samplingRate,  
    int channels,  
    OpusCodec.FrameDuration frameDurationUs,  
    int bitrate,  
    object userdata = null ) [static]
```

Create stream info for an Opus audio stream.

Parameters

<i>samplingRate</i>	Audio sampling rate.
<i>channels</i>	Number of channels.
<i>frameDurationUs</i>	Uncompressed frame (audio packet) size in microseconds.
<i>bitrate</i>	Stream bitrate (in bits/second).
<i>userdata</i>	Optional user data. Should be serializable by Photon .

Returns

[VoiceInfo](#) instance.

4.131.3 Property Documentation

4.131.3.1 Bitrate

```
int Bitrate [get], [set]
```

Target bitrate (in bits/second).

4.131.3.2 Channels

```
int Channels [get], [set]
```

Number of channels.

4.131.3.3 FPS

```
int FPS [get], [set]
```

Video frames per second

4.131.3.4 FrameDurationSamples

```
int FrameDurationSamples [get]
```

Uncompressed frame (data packet) size in samples.

4.131.3.5 FrameDurationUs

```
int FrameDurationUs [get], [set]
```

Uncompressed frame (audio packet) size in microseconds.

4.131.3.6 FrameSize

```
int FrameSize [get]
```

Uncompressed frame (data packet) array size.

4.131.3.7 Height

```
int Height [get], [set]
```

Video height

4.131.3.8 KeyFrameInt

```
int KeyFrameInt [get], [set]
```

Video keyframe interval in frames

4.131.3.9 SamplingRate

```
int SamplingRate [get], [set]
```

Audio sampling rate (frequency, in Hz).

4.131.3.10 UserData

```
object UserData [get], [set]
```

Optional user data. Should be serializable by [Photon](#).

4.131.3.11 Width

```
int Width [get], [set]
```

Video width.

4.132 AudioUtil.VoiceLevelDetectCalibrate< T > Class Template Reference

Utility Audio Processor [Voice](#) Detection Calibration.

Inherits [IProcessor< T >](#).

Public Member Functions

- [VoiceLevelDetectCalibrate](#) (int samplingRate, int channels)
Create new [VoiceLevelDetectCalibrate](#) instance
- void [Calibrate](#) (int durationMs, Action< float > onCalibrated=null)
Start calibration
- T[] [Process](#) (T[] buf)
Process a frame of audio data.
- void **Dispose** ()

Properties

- [ILevelMeter](#) [LevelMeter](#) [get]
The [LevelMeter](#) in use.
- [IVoiceDetector](#) [VoiceDetector](#) [get]
The [VoiceDetector](#) in use
- bool [IsCalibrating](#) [get]

4.132.1 Detailed Description

Utility Audio Processor [Voice](#) Detection Calibration.

Encapsulates level meter, voice detector and voice detector calibrator in single instance.

4.132.2 Constructor & Destructor Documentation

4.132.2.1 VoiceLevelDetectCalibrate()

```
VoiceLevelDetectCalibrate (
    int samplingRate,
    int channels )
```

Create new [VoiceLevelDetectCalibrate](#) instance

Parameters

<i>samplingRate</i>	Sampling rate of the audio signal (in Hz).
<i>numChannels</i>	Number of channels in the audio signal.

4.132.3 Member Function Documentation

4.132.3.1 Calibrate()

```
void Calibrate (
    int durationMs,
    Action< float > onCalibrated = null )
```

Start calibration

Parameters

<i>durationMs</i>	Duration of the calibration procedure (in milliseconds).
<i>onCalibrated</i>	Called when calibration is complete. Parameter is new threshold value.

This activates the Calibration process. It will reset the given [LevelMeter](#)'s AccumAvgPeakAmp (accumulated average peak amplitude), and when the duration has passed, use it for the [VoiceDetector](#)'s detection threshold.

4.132.3.2 Process()

```
T [ ] Process (
    T [ ] buf )
```

Process a frame of audio data.

Parameters

<i>buf</i>	Buffer containing input audio data
------------	------------------------------------

Returns

Buffer containing output audio data or null if frame has been discarded (VAD)

Implements [IProcessor< T >](#).

4.132.4 Property Documentation

4.132.4.1 LevelMeter

```
ILevelMeter LevelMeter [get]
```

The [LevelMeter](#) in use.

4.132.4.2 VoiceDetector

```
IVoiceDetector VoiceDetector [get]
```

The [VoiceDetector](#) in use

4.133 VoiceLogger Class Reference

Inherits [ILogger](#).

Public Member Functions

- **VoiceLogger** (Object context, string tag, DebugLevel level=DebugLevel.ERROR)
- **VoiceLogger** (string tag, DebugLevel level=DebugLevel.ERROR)
- void **LogError** (string fmt, params object[] args)
- void **LogWarning** (string fmt, params object[] args)
- void **LogInfo** (string fmt, params object[] args)
- void **LogDebug** (string fmt, params object[] args)

Properties

- string **Tag** [get, set]
- DebugLevel **LogLevel** [get, set]
- bool **IsErrorEnabled** [get]
- bool **IsWarningEnabled** [get]
- bool **IsInfoEnabled** [get]
- bool **IsDebugEnabled** [get]

4.134 WaveFormat Class Reference

Defines the format of waveform-audio data.

Inherits [ICloneable](#), and [IEquatable< WaveFormat >](#).

Inherited by [WaveFormatExtensible](#).

Public Member Functions

- [WaveFormat](#) ()
Initializes a new instance of the [WaveFormat](#) class with a sample rate of 44100 Hz, bits per sample of 16 bit, 2 channels and PCM as the format type.
- [WaveFormat](#) (int sampleRate, int bits, int channels)
Initializes a new instance of the [WaveFormat](#) class with PCM as the format type.
- [WaveFormat](#) (int sampleRate, int bits, int channels, [AudioEncoding](#) encoding)
Initializes a new instance of the [WaveFormat](#) class.
- [WaveFormat](#) (int sampleRate, int bits, int channels, [AudioEncoding](#) encoding, int extraSize)
Initializes a new instance of the [WaveFormat](#) class.
- long [MillisecondsToBytes](#) (double milliseconds)
Converts a duration in milliseconds to a duration in bytes.
- double [BytesToMilliseconds](#) (long bytes)
Converts a duration in bytes to a duration in milliseconds.
- virtual bool [Equals](#) ([WaveFormat](#) other)
Indicates whether the current object is equal to another object of the same type.
- override string [ToString](#) ()
Returns a string which describes the [WaveFormat](#).
- virtual object [Clone](#) ()
Creates a new [WaveFormat](#) object that is a copy of the current instance.

Protected Member Functions

- virtual void [UpdateProperties](#) ()
Updates the [BlockAlign](#)- and the [BytesPerSecond](#)-property.

Properties

- virtual int [Channels](#) [get, set]
Gets the number of channels in the waveform-audio data. Mono data uses one channel and stereo data uses two channels.
- virtual int [SampleRate](#) [get, set]
Gets the sample rate, in samples per second (hertz).
- virtual int [BytesPerSecond](#) [get, set]
Gets the required average data transfer rate, in bytes per second. For example, 16-bit stereo at 44.1 kHz has an average data rate of 176,400 bytes per second (2 channels — 2 bytes per sample per channel — 44,100 samples per second).
- virtual int [BlockAlign](#) [get, set]
Gets the block alignment, in bytes. The block alignment is the minimum atomic unit of data. For PCM data, the block alignment is the number of bytes used by a single sample, including data for both channels if the data is stereo. For example, the block alignment for 16-bit stereo PCM is 4 bytes (2 channels x 2 bytes per sample).
- virtual int [BitsPerSample](#) [get, set]
Gets the number of bits, used to store one sample.
- virtual int [ExtraSize](#) [get, set]
Gets the size (in bytes) of extra information. This value is mainly used for marshalling.
- virtual int [BytesPerSample](#) [get]
Gets the number of bytes, used to store one sample.
- virtual int [BytesPerBlock](#) [get]
Gets the number of bytes, used to store one block. This value equals [BytesPerSample](#) multiplied with [Channels](#).
- virtual [AudioEncoding WaveFormatTag](#) [get, set]
Gets the waveform-audio format type.

4.134.1 Detailed Description

Defines the format of waveform-audio data.

4.134.2 Constructor & Destructor Documentation

4.134.2.1 WaveFormat() [1/4]

[WaveFormat](#) ()

Initializes a new instance of the [WaveFormat](#) class with a sample rate of 44100 Hz, bits per sample of 16 bit, 2 channels and PCM as the format type.

4.134.2.2 WaveFormat() [2/4]

```
WaveFormat (
    int sampleRate,
    int bits,
    int channels )
```

Initializes a new instance of the [WaveFormat](#) class with PCM as the format type.

Parameters

<i>sampleRate</i>	Samples per second.
<i>bits</i>	Number of bits, used to store one sample.
<i>channels</i>	Number of channels in the waveform-audio data.

4.134.2.3 WaveFormat() [3/4]

```
WaveFormat (  
    int sampleRate,  
    int bits,  
    int channels,  
    AudioEncoding encoding )
```

Initializes a new instance of the [WaveFormat](#) class.

Parameters

<i>sampleRate</i>	Samples per second.
<i>bits</i>	Number of bits, used to store one sample.
<i>channels</i>	Number of channels in the waveform-audio data.
<i>encoding</i>	Format type or encoding of the wave format.

4.134.2.4 WaveFormat() [4/4]

```
WaveFormat (  
    int sampleRate,  
    int bits,  
    int channels,  
    AudioEncoding encoding,  
    int extraSize )
```

Initializes a new instance of the [WaveFormat](#) class.

Parameters

<i>sampleRate</i>	Samples per second.
<i>bits</i>	Number of bits, used to store one sample.
<i>channels</i>	Number of channels in the waveform-audio data.
<i>encoding</i>	Format type or encoding of the wave format.
<i>extraSize</i>	Size (in bytes) of extra information. This value is mainly used for marshalling.

4.134.3 Member Function Documentation

4.134.3.1 BytesToMilliseconds()

```
double BytesToMilliseconds (
    long bytes )
```

Converts a duration in bytes to a duration in milliseconds.

Parameters

<i>bytes</i>	Duration in bytes to convert to a duration in milliseconds.
--------------	---

Returns

Duration in milliseconds.

4.134.3.2 Clone()

```
virtual object Clone ( ) [virtual]
```

Creates a new [WaveFormat](#) object that is a copy of the current instance.

Returns

A copy of the current instance.

Reimplemented in [WaveFormatExtensible](#).

4.134.3.3 Equals()

```
virtual bool Equals (
    WaveFormat other ) [virtual]
```

Indicates whether the current object is equal to another object of the same type.

Parameters

<i>other</i>	The WaveFormat to compare with this WaveFormat .
--------------	--

Returns

true if the current object is equal to the other parameter; otherwise, false.

4.134.3.4 MillisecondsToBytes()

```
long MillisecondsToBytes (
    double milliseconds )
```

Converts a duration in milliseconds to a duration in bytes.

Parameters

<i>milliseconds</i>	Duration in millisecond to convert to a duration in bytes.
---------------------	--

Returns

Duration in bytes.

4.134.3.5 ToString()

```
override string ToString ( )
```

Returns a string which describes the [WaveFormat](#).

Returns

A string which describes the [WaveFormat](#).

4.134.3.6 UpdateProperties()

```
virtual void UpdateProperties ( ) [protected], [virtual]
```

Updates the [BlockAlign](#)- and the [BytesPerSecond](#)-property.

4.134.4 Property Documentation

4.134.4.1 BitsPerSample

```
virtual int BitsPerSample [get], [set]
```

Gets the number of bits, used to store one sample.

4.134.4.2 BlockAlign

```
virtual int BlockAlign [get], [set]
```

Gets the block alignment, in bytes. The block alignment is the minimum atomic unit of data. For PCM data, the block alignment is the number of bytes used by a single sample, including data for both channels if the data is stereo. For example, the block alignment for 16-bit stereo PCM is 4 bytes (2 channels x 2 bytes per sample).

4.134.4.3 BytesPerBlock

```
virtual int BytesPerBlock [get]
```

Gets the number of bytes, used to store one block. This value equals [BytesPerSample](#) multiplied with [Channels](#).

4.134.4.4 BytesPerSample

```
virtual int BytesPerSample [get]
```

Gets the number of bytes, used to store one sample.

4.134.4.5 BytesPerSecond

```
virtual int BytesPerSecond [get], [set]
```

Gets the required average data transfer rate, in bytes per second. For example, 16-bit stereo at 44.1 kHz has an average data rate of 176,400 bytes per second (2 channels — 2 bytes per sample per channel — 44,100 samples per second).

4.134.4.6 Channels

```
virtual int Channels [get], [set]
```

Gets the number of channels in the waveform-audio data. Mono data uses one channel and stereo data uses two channels.

4.134.4.7 ExtraSize

```
virtual int ExtraSize [get], [set]
```

Gets the size (in bytes) of extra information. This value is mainly used for marshalling.

4.134.4.8 SampleRate

```
virtual int SampleRate [get], [set]
```

Gets the sample rate, in samples per second (hertz).

4.134.4.9 WaveFormatTag

```
virtual AudioEncoding WaveFormatTag [get], [set]
```

Gets the waveform-audio format type.

4.135 WaveFormatExtensible Class Reference

Defines the format of waveform-audio data for formats having more than two channels or higher sample resolutions than allowed by [WaveFormat](#). Can be used to define any format that can be defined by [WaveFormat](#). For more information see and .

Inherits [WaveFormat](#).

Public Member Functions

- [WaveFormatExtensible](#) (int sampleRate, int bits, int channels, Guid subFormat)
Initializes a new instance of the [WaveFormatExtensible](#) class.
- [WaveFormatExtensible](#) (int sampleRate, int bits, int channels, Guid subFormat, [ChannelMask](#) channelMask)
Initializes a new instance of the [WaveFormatExtensible](#) class.
- [WaveFormat ToWaveFormat](#) ()
Converts the [WaveFormatExtensible](#) instance to a raw [WaveFormat](#) instance by converting the [SubFormat](#) to the equal [WaveFormat.WaveFormatTag](#).
- override object [Clone](#) ()
Creates a new [WaveFormat](#) object that is a copy of the current instance.
- override string [ToString](#) ()
Returns a string which describes the [WaveFormatExtensible](#).

Static Public Member Functions

- static Guid [SubTypeFromWaveFormat](#) ([WaveFormat](#) waveFormat)
Returns the SubType-Guid of a waveFormat . If the specified waveFormat does not contain a SubType-Guid, the [WaveFormat.WaveFormatTag](#) gets converted to the equal SubType-Guid using the [AudioSubTypes.SubTypeFromEncoding](#) method.

Properties

- int [ValidBitsPerSample](#) [get, protected set]
Gets the number of bits of precision in the signal. Usually equal to [WaveFormat.BitsPerSample](#). However, [WaveFormat.BitsPerSample](#) is the container size and must be a multiple of 8, whereas [ValidBitsPerSample](#) can be any value not exceeding the container size. For example, if the format uses 20-bit samples, [WaveFormat.BitsPerSample](#) must be at least 24, but [ValidBitsPerSample](#) is 20.
- int [SamplesPerBlock](#) [get, protected set]
Gets the number of samples contained in one compressed block of audio data. This value is used in buffer estimation. This value is used with compressed formats that have a fixed number of samples within each block. This value can be set to 0 if a variable number of samples is contained in each block of compressed audio data. In this case, buffer estimation and position information needs to be obtained in other ways.
- [ChannelMask](#) [ChannelMask](#) [get, protected set]
Gets a bitmask specifying the assignment of channels in the stream to speaker positions.
- Guid [SubFormat](#) [get, protected set]
Subformat of the data, such as [AudioSubTypes.Pcm](#). The subformat information is similar to that provided by the tag in the [WaveFormat](#) class's [WaveFormat.WaveFormatTag](#) member.

Additional Inherited Members

4.135.1 Detailed Description

Defines the format of waveform-audio data for formats having more than two channels or higher sample resolutions than allowed by [WaveFormat](#). Can be used to define any format that can be defined by [WaveFormat](#). For more information see and .

4.135.2 Constructor & Destructor Documentation

4.135.2.1 WaveFormatExtensible() [1/2]

```
WaveFormatExtensible (
    int sampleRate,
    int bits,
    int channels,
    Guid subFormat )
```

Initializes a new instance of the [WaveFormatExtensible](#) class.

Parameters

<i>sampleRate</i>	Samplerate of the waveform-audio. This value will get applied to the WaveFormat.SampleRate property.
<i>bits</i>	Bits per sample of the waveform-audio. This value will get applied to the WaveFormat.BitsPerSample property and the ValidBitsPerSample property.
<i>channels</i>	Number of channels of the waveform-audio. This value will get applied to the WaveFormat.Channels property.
<i>subFormat</i>	Subformat of the data. This value will get applied to the SubFormat property.

4.135.2.2 WaveFormatExtensible() [2/2]

```
WaveFormatExtensible (
    int sampleRate,
    int bits,
    int channels,
    Guid subFormat,
    ChannelMask channelMask )
```

Initializes a new instance of the [WaveFormatExtensible](#) class.

Parameters

<i>sampleRate</i>	Samplerate of the waveform-audio. This value will get applied to the WaveFormat.SampleRate property.
<i>bits</i>	Bits per sample of the waveform-audio. This value will get applied to the WaveFormat.BitsPerSample property and the ValidBitsPerSample property.
<i>channels</i>	Number of channels of the waveform-audio. This value will get applied to the WaveFormat.Channels property.
<i>subFormat</i>	Subformat of the data. This value will get applied to the SubFormat property.
<i>channelMask</i>	Bitmask specifying the assignment of channels in the stream to speaker positions. This value will get applied to the ChannelMask property.

4.135.3 Member Function Documentation

4.135.3.1 Clone()

```
override object Clone ( ) [virtual]
```

Creates a new [WaveFormat](#) object that is a copy of the current instance.

Returns

A copy of the current instance.

Reimplemented from [WaveFormat](#).

4.135.3.2 SubTypeFromWaveFormat()

```
static Guid SubTypeFromWaveFormat (
    WaveFormat waveFormat ) [static]
```

Returns the SubType-Guid of a *waveFormat* . If the specified *waveFormat* does not contain a SubType-Guid, the [WaveFormat.WaveFormatTag](#) gets converted to the equal SubType-Guid using the [AudioSubTypes.SubTypeFromEncoding](#) method.

Parameters

<i>waveFormat</i>	WaveFormat which gets used to determine the SubType-Guid.
-------------------	---

Returns

SubType-Guid of the specified *waveFormat* .

4.135.3.3 ToString()

```
override string ToString ( )
```

Returns a string which describes the [WaveFormatExtensible](#).

Returns

A string which describes the [WaveFormatExtensible](#).

4.135.3.4 ToWaveFormat()

```
WaveFormat ToWaveFormat ( )
```

Converts the [WaveFormatExtensible](#) instance to a raw [WaveFormat](#) instance by converting the [SubFormat](#) to the equal [WaveFormat.WaveFormatTag](#).

Returns

A simple [WaveFormat](#) instance.

4.135.4 Property Documentation

4.135.4.1 ChannelMask

```
ChannelMask ChannelMask [get], [protected set]
```

Gets a bitmask specifying the assignment of channels in the stream to speaker positions.

4.135.4.2 SamplesPerBlock

```
int SamplesPerBlock [get], [protected set]
```

Gets the number of samples contained in one compressed block of audio data. This value is used in buffer estimation. This value is used with compressed formats that have a fixed number of samples within each block. This value can be set to 0 if a variable number of samples is contained in each block of compressed audio data. In this case, buffer estimation and position information needs to be obtained in other ways.

4.135.4.3 SubFormat

```
Guid SubFormat [get], [protected set]
```

Subformat of the data, such as [AudioSubTypes.Pcm](#). The subformat information is similar to that provided by the tag in the [WaveFormat](#) class's [WaveFormat.WaveFormatTag](#) member.

4.135.4.4 ValidBitsPerSample

```
int ValidBitsPerSample [get], [protected set]
```

Gets the number of bits of precision in the signal. Usually equal to [WaveFormat.BitsPerSample](#). However, [WaveFormat.BitsPerSample](#) is the container size and must be a multiple of 8, whereas [ValidBitsPerSample](#) can be any value not exceeding the container size. For example, if the format uses 20-bit samples, [WaveFormat.BitsPerSample](#) must be at least 24, but [ValidBitsPerSample](#) is 20.

4.136 WaveWriter Class Reference

Encoder for wave files.

Inherits [IDisposable](#), and [IWriteable](#).

Public Member Functions

- [WaveWriter](#) (string fileName, [WaveFormat](#) waveFormat)
Initializes a new instance of the [WaveWriter](#) class.
- [WaveWriter](#) (Stream stream, [WaveFormat](#) waveFormat)
Initializes a new instance of the [WaveWriter](#) class.
- void [Dispose](#) ()
Disposes the [WaveWriter](#) and writes down the wave header.
- void [WriteSample](#) (float sample)
Encodes a single sample.
- void [WriteSamples](#) (float[] samples, int offset, int count)
Encodes multiple samples.
- void [Write](#) (byte[] buffer, int offset, int count)
Encodes raw data in the form of a byte array.
- void [Write](#) (byte value)
Writes down a single byte.
- void [Write](#) (short value)
Writes down a single 16 bit integer value.
- void [Write](#) (int value)
Writes down a single 32 bit integer value.
- void [Write](#) (float value)
Writes down a single 32 bit float value.

Static Public Member Functions

- static void [WriteToFile](#) (string filename, [IWaveSource](#) source, bool deleteFileIfAlreadyExists, int maxLength=-1)

Writes down all audio data of the [IWaveSource](#) to a file.

Protected Member Functions

- virtual void [Dispose](#) (bool disposing)

Disposes the [WaveWriter](#) and writes down the wave header.

Properties

- bool [IsDisposed](#) [get]

Signals if the object has already been disposed

- bool [IsDisposing](#) [get]

Signals if the object is in a disposing state

4.136.1 Detailed Description

Encoder for wave files.

4.136.2 Constructor & Destructor Documentation

4.136.2.1 WaveWriter() [1/2]

```
WaveWriter (
    string fileName,
    WaveFormat waveFormat )
```

Initializes a new instance of the [WaveWriter](#) class.

Parameters

<i>fileName</i>	Filename of the destination file. This filename should typically end with the .wav extension.
<i>waveFormat</i>	Format of the waveform-audio data. Note that the WaveWriter won't convert any data.

4.136.2.2 WaveWriter() [2/2]

```
WaveWriter (
    Stream stream,
    WaveFormat waveFormat )
```

Initializes a new instance of the [WaveWriter](#) class.

Parameters

<i>stream</i>	Destination stream which should be used to store the
<i>waveFormat</i>	Format of the waveform-audio data. Note that the WaveWriter won't convert any data.

4.136.3 Member Function Documentation

4.136.3.1 Dispose() [1/2]

```
void Dispose ( )
```

Disposes the [WaveWriter](#) and writes down the wave header.

4.136.3.2 Dispose() [2/2]

```
virtual void Dispose (
    bool disposing ) [protected], [virtual]
```

Disposes the [WaveWriter](#) and writes down the wave header.

Parameters

<i>disposing</i>	True to release both managed and unmanaged resources; false to release only unmanaged resources.
------------------	--

4.136.3.3 Write() [1/5]

```
void Write (
    byte value )
```

Writes down a single byte.

Parameters

<i>value</i>	Byte to write down.
--------------	---------------------

4.136.3.4 Write() [2/5]

```
void Write (
    byte[] buffer,
    int offset,
    int count )
```

Encodes raw data in the form of a byte array.

Parameters

<i>buffer</i>	Byte array which contains the data to encode.
<i>offset</i>	Zero-based offset in the <i>buffer</i> .
<i>count</i>	Number of bytes to encode.

Implements [IWriteable](#).

4.136.3.5 Write() [3/5]

```
void Write (
    float value )
```

Writes down a single 32 bit float value.

Parameters

<i>value</i>	Value to write down.
--------------	----------------------

4.136.3.6 Write() [4/5]

```
void Write (
    int value )
```

Writes down a single 32 bit integer value.

Parameters

<i>value</i>	Value to write down.
--------------	----------------------

4.136.3.7 Write() [5/5]

```
void Write (
    short value )
```

Writes down a single 16 bit integer value.

Parameters

<i>value</i>	Value to write down.
--------------	----------------------

4.136.3.8 WriteSample()

```
void WriteSample (
    float sample )
```

Encodes a single sample.

Parameters

<i>sample</i>	The sample to encode.
---------------	-----------------------

4.136.3.9 WriteSamples()

```
void WriteSamples (
    float[] samples,
    int offset,
    int count )
```

Encodes multiple samples.

Parameters

<i>samples</i>	Float array which contains the samples to encode.
<i>offset</i>	Zero-based offset in the <i>samples</i> array.
<i>count</i>	Number of samples to encode.

4.136.3.10 WriteToFile()

```
static void WriteToFile (
    string filename,
    IWaveSource source,
    bool deleteFileIfAlreadyExists,
    int maxlength = -1 ) [static]
```

Writes down all audio data of the [IWaveSource](#) to a file.

Parameters

<i>filename</i>	The filename.
<i>source</i>	The source to write down to the file.
<i>deleteFileIfAlreadyExists</i>	if set to <code>true</code> the file will be overwritten if it already exists.
<i>maxlength</i>	The maximum number of bytes to write. Use -1 to write an infinite number of bytes.

This method is obsolete. Use the `Extensions.WriteToWaveStream` extension instead.

4.136.4 Property Documentation

4.136.4.1 IsDisposed

```
bool IsDisposed [get]
```

Signals if the object has already been disposed

4.136.4.2 IsDisposing

```
bool IsDisposing [get]
```

Signals if the object is in a disposing state

4.137 WebRtcAudioDsp Class Reference

Inherits [VoiceComponent](#).

Public Member Functions

- bool [SetOrSwitchAudioListener](#) (AudioListener audioListener)
Set the AudioListener to be used with this [WebRtcAudioDsp](#)
- bool [SetOrSwitchAudioOutCapture](#) (AudioOutCapture audioOutCapture)
Set the [AudioOutCapture](#) to be used with this [WebRtcAudioDsp](#)

Public Attributes

- bool **AECMobileComfortNoise**

Protected Member Functions

- override void **Awake** ()

Properties

- bool **AEC** [get, set]
- bool **AECMobile** [get, set]
- bool **AecHighPass** [get, set]
- int **ReverseStreamDelayMs** [get, set]
- bool **NoiseSuppression** [get, set]
- bool **HighPass** [get, set]
- bool **Bypass** [get, set]
- bool **AGC** [get, set]
- int **AgcCompressionGain** [get, set]
- bool **VAD** [get, set]
- bool **ForceNormalAecInMobile** [get, set]

Additional Inherited Members

4.137.1 Member Function Documentation

4.137.1.1 SetOrSwitchAudioListener()

```
bool SetOrSwitchAudioListener (
    AudioListener audioListener )
```

Set the AudioListener to be used with this [WebRtcAudioDsp](#)

Parameters

<i>audioListener</i>	The audioListener to be used
----------------------	------------------------------

Returns

Success or failure

4.137.1.2 SetOrSwitchAudioOutCapture()

```
bool SetOrSwitchAudioOutCapture (
    AudioOutCapture audioOutCapture )
```

Set the [AudioOutCapture](#) to be used with this [WebRtcAudioDsp](#)

Parameters

<i>audioOutCapture</i>	The audioOutCapture to be used
------------------------	--------------------------------

Returns

Success or failure

4.138 WebRTCAudioLib Class Reference

Inherited by [WebRTCAudioProcessor](#).

Public Types

- enum **Error**
- enum **Param**

Public Member Functions

- static IntPtr **webrtc_audio_processor_create** (int samplingRate, int channels, int frameSize, int rev↔ SamplingRate, int revChannels)
- static int **webrtc_audio_processor_init** (IntPtr proc)
- static int **webrtc_audio_processor_set_param** (IntPtr proc, int param, int v)
- static int **webrtc_audio_processor_process** (IntPtr proc, short[] buffer, int offset, out bool voiceDetected)
- static int **webrtc_audio_processor_process_reverse** (IntPtr proc, short[] buffer, int bufferSize)
- static void **webrtc_audio_processor_destroy** (IntPtr proc)

4.139 WebRTCAudioProcessor Class Reference

Inherits [WebRTCAudioLib](#), and [IProcessor< short >](#).

Public Member Functions

- **WebRTCAudioProcessor** ([ILogger](#) logger, int frameSize, int samplingRate, int channels, int reverse↔ SamplingRate, int reverseChannels)
- short[] **Process** (short[] buf)
- void **OnAudioOutFrameFloat** (float[] data)
- void **Dispose** ()

Static Public Attributes

- static readonly int[] **SupportedSamplingRates** = { 8000, 16000, 32000, 48000 }

Properties

- int **AECStreamDelayMs** [set]
- bool?? **AEC** [set]
- bool? **AECHighPass** [set]
- bool?? **AECMobile** [set]
- bool? **HighPass** [set]
- bool? **NoiseSuppression** [set]
- bool? **AGC** [set]
- int **AGCCompressionGain** [set]
- int **AGCTargetLevel** [set]
- bool? **AGC2** [set]
- bool? **VAD** [set]
- bool **Bypass** [set]

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