

Photon Voice
v2.3

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Chapter 1

Photon Voice Doxygen Readme

Offline Docs

Manual Generation

To manually generate doxygen offline files:

```
"doxygen .\Docs\doxygen\voice-doxxygen-offline.config"
```

Automatic Generation

Simply run "Docs\generate_offline.bat". Open the file and edit DOXYGEN_PATH accordingly. Also you need a LaTeX distribution installed and some packages/dependencies.

This script will also copy the offline files to their respective locations and then clean up.

Files

HTML

It is not possible to disable HTML files generation. So those are just ignored or cleaned up after generation.

CHM

"PhotonVoice-Documentation.chm" should be copied

to "Assets\Photon\PhotonVoice-Documentation.chm"

from "Docs\TempOutputDocs\VOICE_OFFLINE_HTML\PhotonVoice-Documentation.chm".

PDF

"PhotonVoice-Documentation.pdf" should be copied

to "Assets\Photon\PhotonVoice-Documentation.pdf"

from "Docs\TempOutputDocs\latex\refman.pdf".

Online Docs

To manually generate doxygen online files:

```
"doxygen .\Docs\doxygen\voice-doxxygen-online.config"
```


Chapter 2

Namespace Documentation

2.1 Photon Namespace Reference

Namespaces

- namespace [Voice](#)

2.2 Photon.Voice Namespace Reference

Namespaces

- namespace [IOS](#)
- namespace [PUN](#)
- namespace [Unity](#)

Classes

- class [AudioDesc](#)
- class [AudioInEnumerator](#)
- class [AudioStreamPlayer](#)
- 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 [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 returnig the same array instance as long as it requested with the same array length. If length changes, new array instance created.

- 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 media decoder interface.

- interface [IDecoderDirect](#)

Interface for a media decoder that synchronously decodes data.

- interface [IDecoderQueued](#)

Interface for a media decoder that feeds its data output into a separate method or callback asynchronously, or does not produce output at all.

- interface [IDecoderQueuedOutputImageNative](#)

- interface [IEncoder](#)

Generic media encoder interface.

- interface [IEncoderDataFlow](#)

Interface for a generic media encoder data flow.

- interface [IEncoderDataFlowDirect](#)

Interface for an encoder data flow that synchronously encodes data.

- interface [IEncoderNativeImageDirect](#)

- interface [IEncoderQueued](#)

Interface for an encoder data flow that returns compressed data independently (produces output on its own) or asynchronously (usually from a queue).

- interface [ILocalVoiceAudio](#)

Interface for an outgoing audio stream.

- interface [ILogger](#)

- class [ImageBufferInfo](#)

- class [ImageBufferNative](#)

- class [ImageBufferNativeAlloc](#)

- class [ImageBufferNativeGCHandleSinglePlane](#)

- class [ImageBufferNativePool](#)

- interface [IProcessor](#)

Audio Processor interface.

- interface [IServiceable](#)

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

- interface [ISyncAudioOut](#)

- interface [IVoiceFrontend](#)

- class [LoadBalancingFrontend](#)

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

- 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](#)*
- 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 [PrimitiveArrayPool](#)
 - Pool of Arrays with components of type T, with [ObjectPool](#) info being the array's size.*
- 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 [SpeexLib](#)
- class [SpeexProcessor](#)
- class [UnsupportedCodecException](#)
 - Exception thrown if an unsupported codec is encountered.*
- class [UnsupportedSampleTypeException](#)
 - Exception thrown if an unsupported audio sample type is encountered.*
- class [VoiceClient](#)
 - Base class for [Voice](#) clients implamantations*
- class **VoiceCodec**
- class [VoiceEventCode](#)
 - PhotonVoice communication uses a single type of event, but differentiates transmission Channels by encoding a channelId into [VoiceEventCode](#).*
- struct [VoiceInfo](#)
 - Describes stream properties.*
- class [WebRTCAudioLib](#)
- class [WebRTCAudioProcessor](#)

Enumerations

- enum **EventSubcode** : byte
- enum **EventParam** : byte
- enum [Codec](#)
 - Enum for Media Codecs supported by PhotonVoice.*
- enum **ImageFormat**
- enum **Rotation**
- enum **Flip**

Functions

- delegate void **OnImageOutputNative** (IntPtr buf, int width, int height, int stride)

2.2.1 Enumeration Type Documentation

2.2.1.1 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

2.3 Photon.Voice.IOS Namespace Reference

Classes

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

Enumerations

- enum **AudioSessionCategory**
- enum **AudioSessionMode**
- enum **AudioSessionCategoryOption**

2.4 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.*

2.5 Photon.Voice.Unity Namespace Reference

Namespaces

- namespace [UtilityScripts](#)

Classes

- class [AudioClipWrapper](#)
- class [AudioOutCapture](#)
- interface [ILoggable](#)
- class [IOSAudioForceToSpeaker](#)
- class [Logger](#)
- class [MicWrapper](#)

- 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 [VoiceComponent](#)
- class [VoiceConnection](#)
Component that represents a client voice connection to [Photon](#) Servers.
- class [VoiceLogger](#)
- class [WebRtcAudioDsp](#)

2.6 Photon.Voice.Unity.UtilityScripts Namespace Reference

Classes

- class [ConnectAndJoin](#)
- class [PhotonVoiceLagSimulationGui](#)
- class [TestTone](#)
- class [ToneAudioReader](#)

2.7 POpusCodec Namespace Reference

Namespaces

- namespace [Enums](#)

Classes

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

2.8 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

2.8.1 Enumeration Type Documentation

2.8.1.1 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)

2.8.1.2 enum **Channels** : int [strong]

Enumerator

Mono 1 Channel
Stereo 2 Channels

2.8.1.3 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

2.8.1.4 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.

2.8.1.5 enum **SignalHint** : int [strong]

Enumerator

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

Chapter 3

Class Documentation

3.1 SpeexProcessor.AECLatencyResultType Struct Reference

Public Attributes

- int **LatencyMs**
- int **LatencyDelayedMs**
- bool **PlayDetected**
- bool **PlayDelayedDetected**
- bool **RecDetected**

3.2 AudioClipWrapper Class Reference

Inherits [IAudioReader< T >](#).

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]

3.3 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]

3.4 AudioInEnumerator Class Reference

Inherits IDisposable.

Public Member Functions

- **AudioInEnumerator** (Voice.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]

3.5 AudioOutCapture Class Reference

Inherits MonoBehaviour.

Events

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

3.6 AudioSessionParameters Struct Reference

Public Member Functions

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

Public Attributes

- AudioSessionCategory **Category**
- AudioSessionMode **Mode**
- AudioSessionCategoryOption[] **CategoryOptions**

3.7 AudioSessionParametersPresets Class Reference

Static Public Attributes

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

3.7.1 Member Data Documentation

3.7.1.1 AudioSessionParameters Game [static]

Initial value:

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

3.7.1.2 AudioSessionParameters VoIP [static]

Initial value:

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

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

3.8 AudioStreamPlayer Class Reference

Inherits [IAudioOut](#).

Public Member Functions

- **AudioStreamPlayer** ([Photon.Voice.ILogger](#) logger, [Photon.Voice.ISyncAudioOut](#) audioOut, string logPrefix, bool debugInfo)
- void **Start** (int frequency, int channels, int frameSamples, int playDelayMs)
- void **Service** ()
- void **Push** (float[] frame)
- void **Stop** ()

Properties

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

3.9 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 [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.

3.9.1 Detailed Description

Collection of Audio Utility functions and classes.

3.9.2 Member Function Documentation

3.9.2.1 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.

3.9.2.2 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.

3.9.2.3 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 <i>src</i> .

3.9.2.4 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.

3.9.2.5 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.

3.9.2.6 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.

3.10 BufferReaderPushAdapter< T > Class Template Reference

Simple [BufferReaderPushAdapterBase](#) implementation using a single buffer, using synchronous LocalVoice.Push↔Data

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**

3.10.1 Detailed Description

Simple [BufferReaderPushAdapterBase](#) implementation using a single buffer, using synchronous LocalVoice.Push↔Data

3.10.2 Constructor & Destructor Documentation

3.10.2.1 [BufferReaderPushAdapter](#) (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.

3.10.3 Member Function Documentation

3.10.3.1 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 >](#).

3.11 [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

3.11.1 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.

3.11.2 Constructor & Destructor Documentation

3.11.2.1 [BufferedReaderPushAdapterAsyncPool](#) ([LocalVoice](#) *localVoice*, [IDataReader](#)< T > *reader*)

Create a new [BufferedReaderPushAdapter](#) instance

Parameters

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

3.11.3 Member Function Documentation

3.11.3.1 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 >](#).

3.12 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**

3.12.1 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 cycle at the cost of a buffer copy. Expects localVoice to be a [LocalVoiceFramed<T>](#) of same T.

3.12.2 Constructor & Destructor Documentation

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

Create a new [BufferReaderPushAdapter](#) instance

Parameters

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

3.12.3 Member Function Documentation

3.12.3.1 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 >](#).

3.13 [BufferReaderPushAdapterAsyncPoolFloatToShort](#) Class Reference

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

Inherits [BufferReaderPushAdapterBase< T >](#).

Public Member Functions

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

Additional Inherited Members

3.13.1 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.

3.13.2 Constructor & Destructor Documentation

3.13.2.1 [BufferReaderPushAdapterAsyncPoolFloatToShort](#) ([Voice.LocalVoice](#) localVoice, [Voice.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.

3.13.3 Member Function Documentation

3.13.3.1 override void [Service](#) ([Voice.LocalVoice](#) localVoice)

Do the actual data read/push.

Parameters

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

3.14 [BufferReaderPushAdapterBase](#)< T > Class Template Reference

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

Inherits [IServiceable](#).

Inherited by [BufferReaderPushAdapter< T >](#), [BufferReaderPushAdapterAsyncPool< T >](#), [BufferReaderPushAdapterAsyncPoolCopy< T >](#), and [BufferReaderPushAdapterAsyncPoolFloatToShort](#).

Public Member Functions

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

Protected Attributes

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

3.14.1 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.

3.14.2 Constructor & Destructor Documentation

3.14.2.1 [BufferReaderPushAdapterBase](#) ([IDataReader](#)< T > *reader*)

Create a new [BufferReaderPushAdapterBase](#) instance

Parameters

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

3.14.3 Member Function Documentation

3.14.3.1 void [Dispose](#) ()

Release resources associated with this instance.

3.14.3.2 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 [BufferReaderPushAdapterAsyncPoolCopy< T >](#), [BufferReaderPushAdapterAsyncPool< T >](#), and [BufferReaderPushAdapter< T >](#).

3.15 WebRTCAudioLib.ConfigParam Struct Reference

Public Attributes

- const int **AEC_DELAY_AGNOSTIC** = 12
- const int **AEC_EXTENDED_FILTER** = 13
- const int **AGC_EXPERIMENTAL** = 53
- const int **AGC_EXPERIMENTAL_STARTUP_MIN_VOLUME** = 54
- const int **AGC_EXPERIMENTAL_CLIP_LEVEL_MIN** = 55

3.16 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]

3.17 OpusCodec.Decoder Class Reference

Inherits [IDecoderDirect](#).

Public Member Functions

- **Decoder** ([ILogger](#) logger)
- void **Open** ([VoiceInfo](#) i)
Open (initialize) the decoder.
- byte[] **DecodeToByte** (byte[] buf)
Decode the given raw data buffer.
- float[] **DecodeToFloat** (byte[] buf)
Decode the given raw data buffer to floating point audio.
- short[] **DecodeToShort** (byte[] buf)
Decode the given raw data buffer to 'short' (16-bit) audio.
- void **Dispose** ()

Properties

- string **Error** [[get](#)]

3.17.1 Member Function Documentation

3.17.1.1 byte[] DecodeToByte (byte[] buf)

Decode the given raw data buffer.

Parameters

<i>buf</i>	Buffer of encoded (compressed) data.
------------	--------------------------------------

Returns

Buffer of decoded (uncompressed) data.

Implements [IDecoderDirect](#).

3.17.1.2 float[] DecodeToFloat (byte[] buf)

Decode the given raw data buffer to floating point audio.

Only sensible for audio data.

Parameters

<i>buf</i>	Buffer of encoded (compressed) data.
------------	--------------------------------------

Returns

Buffer of decoded (uncompressed) data.

Implements [IDecoderDirect](#).

3.17.1.3 short[] DecodeToShort (byte[] buf)

Decode the given raw data buffer to 'short' (16-bit) audio.

Only sensible for audio data.

Parameters

<i>buf</i>	Buffer of encoded (compressed) data.
------------	--------------------------------------

Returns

Buffer of decoded (uncompressed) data.

Implements [IDecoderDirect](#).

3.17.1.4 void Open (VoicelInfo info)

Open (initialize) the decoder.

Parameters

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

Implements [IDecoder](#).

3.18 OpusCodec.Encoder< T > Class Template Reference

Inherits [IEncoderDataFlowDirect< T >](#).

Inherited by [OpusCodec.EncoderFloat](#), and [OpusCodec.EncoderShort](#).

Public Member Functions

- void **Dispose** ()
- abstract ArraySegment< byte > [EncodeAndGetOutput](#) (T[] buf)
Encode the given uncompressed media data.

Protected Member Functions

- **Encoder** (VoicelInfo i, [ILogger](#) logger)

Protected Attributes

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

Properties

- string **Error** [get]

3.18.1 Member Function Documentation

3.18.1.1 abstract ArraySegment<byte> EncodeAndGetOutput (T[] buf) [pure virtual]

Encode the given uncompressed media data.

Parameters

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

Returns

Encoded (compressed) data.

Implements [IEncoderDataFlowDirect< T >](#).

3.19 OpusCodec.EncoderFactory Class Reference

Static Public Member Functions

- static [IEncoder](#) **Create** ([VoiceInfo](#) i, [LocalVoice](#) v)

3.20 OpusCodec.EncoderFloat Class Reference

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

Public Member Functions

- override [ArraySegment< byte >](#) **EncodeAndGetOutput** ([float\[\]](#) buf)

Additional Inherited Members

3.21 OpusCodec.EncoderShort Class Reference

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

Public Member Functions

- override [ArraySegment< byte >](#) **EncodeAndGetOutput** ([short\[\]](#) buf)

Additional Inherited Members

3.22 FactoryPrimitiveArrayPool< T > Class Template Reference

[PrimitiveArrayPool<T>](#) as wrapped in object factory interface.

Inherits [ObjectFactory< TType, TInfo >](#).

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]

3.22.1 Detailed Description

PrimitiveArrayPool<T> as wrapped in object factory interface.

Template Parameters

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

3.23 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< TType, TInfo >](#).

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]

3.23.1 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.
----------	---------------------

3.24 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.

3.24.1 Detailed Description

Utility class to re-frame audio packets.

3.24.2 Constructor & Destructor Documentation

3.24.2.1 `Framer (int frameSize)`

Create new `Framer` instance.

3.24.3 Member Function Documentation

3.24.3.1 `int Count (int bufLen)`

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

Parameters

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

Returns

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

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

Append arbitrary-sized buffer and return available full frames.

Parameters

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

Returns

Enumerator of full frames (might be none).

3.25 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.

3.25.1 Detailed Description

Audio Source interface.

3.25.2 Property Documentation

3.25.2.1 `int Channels` [get]

Number of channels in the audio signal.

3.25.2.2 `string Error` [get]

If not null, audio object is in invalid state.

3.25.2.3 `int SamplingRate` [get]

Sampling rate of the audio signal (in Hz).

3.26 IAudioOut Interface Reference

Inherited by [AudioStreamPlayer](#), and [ISyncAudioOut](#).

Public Member Functions

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

Properties

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

3.27 IAudioPusher< T > Interface Template Reference

Audio Pusher interface.

Inherits [IAudioDesc](#).

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

Public Member Functions

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

Additional Inherited Members

3.27.1 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,

3.27.2 Member Function Documentation

3.27.2.1 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 >](#).

3.28 IAudioReader< T > Interface Template Reference

Audio Reader interface.

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

Inherited by [AudioUtil.ToneAudioReader< T >](#), [AudioClipWrapper](#), [MicWrapper](#), and [ToneAudioReader](#).

Additional Inherited Members

3.28.1 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).

3.29 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.

3.29.1 Detailed Description

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

3.29.2 Member Function Documentation

3.29.2.1 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 >](#).

3.30 IDecoder Interface Reference

Generic media decoder interface.

Inherits IDisposable.

Inherited by [IDecoderDirect](#), and [IDecoderQueued](#).

Public Member Functions

- void [Open](#) ([VoiceInfo](#) info)
Open (initialize) the decoder.

Properties

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

3.30.1 Detailed Description

Generic media decoder interface.

3.30.2 Member Function Documentation

3.30.2.1 void Open ([VoiceInfo](#) info)

Open (initialize) the decoder.

Parameters

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

Implemented in [OpusCodec.Decoder](#).

3.30.3 Property Documentation

3.30.3.1 string Error [get]

If not null, the object is in invalid state.

3.31 IDecoderDirect Interface Reference

Interface for a media decoder that synchronously decodes data.

Inherits [IDecoder](#).

Inherited by [OpusCodec.Decoder](#).

Public Member Functions

- `byte[] DecodeToByte (byte[] buf)`
Decode the given raw data buffer.
- `float[] DecodeToFloat (byte[] buf)`
Decode the given raw data buffer to floating point audio.
- `short[] DecodeToShort (byte[] buf)`
Decode the given raw data buffer to 'short' (16-bit) audio.

Additional Inherited Members

3.31.1 Detailed Description

Interface for a media decoder that synchronously decodes data.

3.31.2 Member Function Documentation

3.31.2.1 `byte[] DecodeToByte (byte[] buf)`

Decode the given raw data buffer.

Parameters

<i>buf</i>	Buffer of encoded (compressed) data.
------------	--------------------------------------

Returns

Buffer of decoded (uncompressed) data.

Implemented in [OpusCodec.Decoder](#).

3.31.2.2 `float[] DecodeToFloat (byte[] buf)`

Decode the given raw data buffer to floating point audio.

Only sensible for audio data.

Parameters

<i>buf</i>	Buffer of encoded (compressed) data.
------------	--------------------------------------

Returns

Buffer of decoded (uncompressed) data.

Implemented in [OpusCodec.Decoder](#).

3.31.2.3 short [] DecodeToShort (byte[] buf)

Decode the given raw data buffer to 'short' (16-bit) audio.

Only sensible for audio data.

Parameters

<i>buf</i>	Buffer of encoded (compressed) data.
------------	--------------------------------------

Returns

Buffer of decoded (uncompressed) data.

Implemented in [OpusCodec.Decoder](#).

3.32 IDecoderQueued Interface Reference

Interface for a media decoder that feeds its data output into a separate method or callback asynchronously, or does not produce output at all.

Inherits [IDecoder](#).

Inherited by [IDecoderQueuedOutputImageNative](#).

Public Member Functions

- void [Decode](#) (byte[] buf)
Decode the given raw data buffer.

Additional Inherited Members

3.32.1 Detailed Description

Interface for a media decoder that feeds its data output into a separate method or callback asynchronously, or does not produce output at all.

3.32.2 Member Function Documentation

3.32.2.1 void Decode (byte[] buf)

Decode the given raw data buffer.

This function will be called also for every missing frame, with buf = null.

Parameters

<i>buf</i>	Buffer of encoded (compressed) data.
------------	--------------------------------------

3.33 IDecoderQueuedOutputImageNative Interface Reference

Inherits [IDecoderQueued](#).

Properties

- ImageFormat **OutputImageFormat** [get, set]
- Flip **OutputImageFlip** [get, set]
- Func< int, int, IntPtr > **OutputImageBufferGetter** [get, set]
- OnImageOutputNative **OnOutputImage** [get, set]

Additional Inherited Members

3.34 IEncoder Interface Reference

Generic media encoder interface.

Inherits IDisposable.

Inherited by [IEncoderDataFlow< T >](#), [IEncoderNativeImageDirect](#), and [IEncoderQueued](#).

Properties

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

3.34.1 Detailed Description

Generic media encoder interface.

3.34.2 Property Documentation

3.34.2.1 string Error [get]

If not null, the object is in invalid state.

3.35 IEncoderDataFlow< T > Interface Template Reference

Interface for a generic media encoder data flow.

Inherits [IEncoder](#).

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

Additional Inherited Members

3.35.1 Detailed Description

Interface for a generic media encoder data flow.

3.36 IEncoderDataFlowDirect< T > Interface Template Reference

Interface for an encoder data flow that synchronously encodes data.

Inherits [IEncoderDataFlow< T >](#).

Inherited by [OpusCodec.Encoder< T >](#).

Public Member Functions

- `ArraySegment< byte > EncodeAndGetOutput (T[] buf)`
Encode the given uncompressed media data.

Additional Inherited Members

3.36.1 Detailed Description

Interface for an encoder data flow that synchronously encodes data.

3.36.2 Member Function Documentation

3.36.2.1 `ArraySegment<byte> EncodeAndGetOutput (T[] buf)`

Encode the given uncompressed media data.

Parameters

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

Returns

Encoded (compressed) data.

Implemented in [OpusCodec.Encoder< T >](#).

3.37 IEncoderNativeImageDirect Interface Reference

Inherits [IEncoder](#).

Public Member Functions

- `IEnumerable< ArraySegment< byte > > EncodeAndGetOutput (IntPtr[] buf, int width, int height, int[] stride, ImageFormat imageFormat, Rotation rotation, Flip flip)`

Additional Inherited Members

3.38 IEncoderQueued Interface Reference

Interface for an encoder data flow that returns compressed data independently (produces output on its own) or asynchronously (usually from a queue).

Inherits [IEncoder](#).

Public Member Functions

- `IEnumerable< ArraySegment< byte > > GetOutput ()`
Get an Enumerable of buffers containing encoded (compressed) data.

Additional Inherited Members

3.38.1 Detailed Description

Interface for an encoder data flow that returns compressed data independently (produces output on its own) or asynchronously (usually from a queue).

3.38.2 Member Function Documentation

3.38.2.1 `IEnumerable<ArraySegment<byte>> GetOutput ()`

Get an Enumerable of buffers containing encoded (compressed) data.

Returns

Encoded (compressed) data.

3.39 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.

3.39.1 Detailed Description

Audio Level Metering interface.

3.39.2 Member Function Documentation

3.39.2.1 `void ResetAccumAvgPeakAmp ()`

Reset [AccumAvgPeakAmp](#).

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

3.39.3 Property Documentation

3.39.3.1 float AccumAvgPeakAmp [get]

Average of CurrentPeakAmps since last reset.

3.39.3.2 float CurrentAvgAmp [get]

Average amplitude value over last half second.

3.39.3.3 float CurrentPeakAmp [get]

Maximum amplitude value over last half second sec.

3.40 ILocalVoiceAudio Interface Reference

Interface for an outgoing audio stream.

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

Public Member Functions

- void [VoiceDetectorCalibrate](#) (int durationMs)
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.

3.40.1 Detailed Description

Interface for an outgoing audio stream.

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

3.40.2 Member Function Documentation

3.40.2.1 void VoiceDetectorCalibrate (int durationMs)

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).
-------------------	--

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

3.40.3 Property Documentation

3.40.3.1 [AudioUtil.ILevelMeter](#) **LevelMeter** [get]

The LevelMeter utility in use.

3.40.3.2 [AudioUtil.IVoiceDetector](#) **VoiceDetector** [get]

The VoiceDetector in use.

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

3.40.3.3 **bool** [VoiceDetectorCalibrating](#) [get]

If true, voice detector calibration is in progress.

3.41 ILoggable Interface Reference

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

Properties

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

3.42 ILogger Interface Reference

Inherited by [IVoiceFrontend](#), [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)

3.43 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]

3.44 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]

3.45 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

3.46 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

3.47 ImageBufferNativePool< T > Class Template Reference

Inherits [ObjectPool< TType, TInfo >](#).

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

3.48 IOSAudioForceToSpeaker Class Reference

Inherits MonoBehaviour.

3.49 IProcessor< T > Interface Template Reference

Audio Processor interface.

Inherits IDisposable.

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

Public Member Functions

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

3.49.1 Detailed Description

Audio Processor interface.

3.49.2 Member Function Documentation

3.49.2.1 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

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

3.50 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.

3.50.1 Detailed Description

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

3.50.2 Member Function Documentation

3.50.2.1 void Service ([LocalVoice](#) localVoice)

Service function that should be called regularly.

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

3.51 ISyncAudioOut Interface Reference

Inherits [IAudioOut](#).

Inherited by [UnityAudioOut](#).

Public Member Functions

- void **Pause** ()
- void **UnPause** ()

Properties

- int **PlaySamplePos** [get, set]

3.52 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.

3.52.1 Detailed Description

Voice Activity Detector interface.

3.52.2 Property Documentation

3.52.2.1 int ActivityDelayMs [get], [set]

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

3.52.2.2 bool Detected [get]

If true, voice detected.

3.52.2.3 DateTime DetectedTime [get]

Last time when switched to detected state.

3.52.2.4 bool On [get], [set]

If true, voice detection enabled.

3.52.2.5 float Threshold [get], [set]

Voice detected as soon as signal level exceeds threshold.

3.52.3 Event Documentation

3.52.3.1 Action OnDetected

Called when switched to detected state.

3.53 IVoiceFrontend Interface Reference

Inherits [ILogger](#).

Inherited by [LoadBalancingFrontend](#).

Public Member Functions

- int **AssignChannel** ([VoiceInfo](#) v)
- 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, byte evNumber, byte voiceId, int channelId, [LocalVoice](#) localVoice)
- string **ChannelIdStr** (int channelId)
- string **PlayerIdStr** (int playerId)
- void **SetDebugEchoMode** ([LocalVoice](#) v)

3.54 AudioUtil.LevelMeter< T > Class Template Reference

Audio Level Meter.

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

Inherited by [AudioUtil.LevelMeterFloat](#), and [AudioUtil.LevelMeterShort](#).

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[] **buffer**
- int **prevValuesPtr**
- float **accumAvgPeakAmpSum**
- int **accumAvgPeakAmpCount**

Properties

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

3.54.1 Detailed Description

Audio Level Meter.

3.54.2 Member Function Documentation

3.54.2.1 `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

Implements [IProcessor< T >](#).

3.54.2.2 `void ResetAccumAvgPeakAmp ()`

Reset AccumAvgPeakAmp.

Implements [AudioUtil.ILevelMeter](#).

3.55 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]

3.55.1 Detailed Description

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

3.55.2 Member Function Documentation

3.55.2.1 void ResetAccumAvgPeakAmp ()

Reset AccumAvgPeakAmp.

Implements [AudioUtil.LevelMeter](#).

3.56 AudioUtil.LevelMeterFloat Class Reference

[LevelMeter](#) specialization for float audio.

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

Public Member Functions

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

Additional Inherited Members

3.56.1 Detailed Description

[LevelMeter](#) specialization for float audio.

3.56.2 Constructor & Destructor Documentation

3.56.2.1 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.

3.57 AudioUtil.LevelMeterShort Class Reference

[LevelMeter](#) specialization for short audio.

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

Public Member Functions

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

Additional Inherited Members

3.57.1 Detailed Description

[LevelMeter](#) specialization for short audio.

3.57.2 Constructor & Destructor Documentation

3.57.2.1 [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.

3.58 LoadBalancingFrontend Class Reference

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

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

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)
- int **AssignChannel** ([VoiceInfo](#) v)
- bool **IsChannelJoined** (int channelId)
- void **SetDebugEchoMode** ([LocalVoice](#) v)
- [LoadBalancingFrontend](#) (ConnectionProtocol connectionProtocol=ConnectionProtocol.Udp)
Initializes a new [LoadBalancingFrontend](#).
- 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)
Change audio groups listened by client. Works only while joined to a voice room.
- void **SendVoicesInfo** (IEnumerable< [LocalVoice](#) > voices, int channelId, int targetPlayerId)
- void [SendDebugEchoVoicesInfo](#) (int channelId)
Send VoicesInfo events to the local player for all voices that have DebugEcho enabled.
- void **SendVoiceRemove** ([LocalVoice](#) voice, int channelId, int targetPlayerId)
- void **SendFrame** (ArraySegment< byte > data, byte evNumber, byte voiceld, int channelId, [LocalVoice](#) localVoice)
- string **ChannelIdStr** (int channelId)
- string **PlayerIdStr** (int playerId)
- void [Dispose](#) ()
Releases all resources used by the [LoadBalancingFrontend](#) instance.

Protected Attributes

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

Properties

- [VoiceClient](#) [VoiceClient](#) [get]

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

- byte [GlobalAudioGroup](#) [get, set]

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

3.58.1 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 .

3.58.2 Constructor & Destructor Documentation

3.58.2.1 [LoadBalancingFrontend](#) ([ConnectionProtocol](#) *connectionProtocol* = [ConnectionProtocol.Udp](#))

Initializes a new [LoadBalancingFrontend](#).

Parameters

<i>connection↔ Protocol</i>	Connection protocol (UDP or TCP). ConnectionProtocol
---------------------------------	--

3.58.3 Member Function Documentation

3.58.3.1 virtual bool [ChangeAudioGroups](#) (byte[] *groupsToRemove*, byte[] *groupsToAdd*) [virtual]

Change audio groups listened by client. Works only while joined to a voice room.

[LocalVoice.Group](#) [SetGlobalAudioGroup](#)(byte)

Note the difference between passing null and byte[0]: null won't add/remove any groups. byte[0] will add/remove all (existing) groups. First, removing groups is executed. This way, you could leave all groups and join only the ones provided.

Parameters

<i>groupsTo↔ Remove</i>	Groups to remove from listened. Null will not leave any. A byte[0] will remove all.
<i>groupsToAdd</i>	Groups to add to listened. Null will not add any. A byte[0] will add all current.

Returns

If request could be enqueued for sending

3.58.3.2 void Dispose ()

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

3.58.3.3 void SendDebugEchoVoicesInfo (int *channelId*)

Send VoicesInfo events to the local player for all voices that have DebugEcho enabled.

This function will call SendVoicesInfo for all local voices of our [VoiceClient](#) that have DebugEchoMode set to true, with the given channel ID, and the local Player's ActorNumber as target.

Parameters

<i>channelId</i>	Transport Channel ID
------------------	----------------------

3.58.3.4 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).

3.58.4 Property Documentation

3.58.4.1 byte GlobalAudioGroup [get], [set]

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

[LocalVoice.Group](#) [ChangeAudioGroups\(byte\[\], byte\[\]\)](#)

3.58.4.2 VoiceClient VoiceClient [get]

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

3.59 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** ()

Public Attributes

- const int **DATA_POOL_CAPACITY** = 50

Protected Member Functions

- void **resetNoTransmitCnt** ()

Protected Attributes

- [IEncoder](#) **encoder**
- [VoiceClient](#) **voiceClient**
- volatile bool **disposed**
- object **disposeLock** = new object()

Properties

- byte [Group](#) [get, set]
If AudioGroup != 0, voice's data is sent only to clients listening to this group.
- [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 frontend.

3.59.1 Detailed Description

Represents outgoing data stream.

3.59.2 Member Function Documentation

3.59.2.1 void RemoveSelf ()

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

.

3.59.3 Property Documentation

3.59.3.1 `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 frontend.

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

Send data encrypted.

3.59.3.3 `int FramesSent` `[get]`

Sent frames counter.

3.59.3.4 `int FramesSentBytes` `[get]`

Sent frames bytes counter.

3.59.3.5 `byte Group` `[get]`, `[set]`

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

[LoadBalancingFrontend.ChangeAudioGroups\(byte\[\], byte\[\]\)](#)

3.59.3.6 `VoiceInfo Info` `[get]`

Returns Info structure assigned on local voice cration.

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

Returns true if stream broadcasts.

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

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

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

Send data reliable.

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

If true, stream data broadcasted.

3.60 `LocalVoiceAudio< T >` Class Template Reference

Outgoing audio stream.

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

Inherited by [LocalVoiceAudioFloat](#), and [LocalVoiceAudioShort](#).

Public Member Functions

- void [VoiceDetectorCalibrate](#) (int durationMs)
Trigger voice detector calibration process.

Static Public Member Functions

- static [LocalVoiceAudio< T >](#) [Create](#) ([VoiceClient](#) voiceClient, byte voiceld, [IEncoder](#) encoder, [VoiceInfo](#) voiceInfo, 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**
- int **sourceSamplingRateHz**
- 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

3.60.1 Detailed Description

Outgoing audio stream.

3.60.2 Member Function Documentation

- 3.60.2.1 static [LocalVoiceAudio<T>](#) [Create](#) ([VoiceClient](#) voiceClient, byte voiceld, [IEncoder](#) encoder, [VoiceInfo](#) voiceInfo, 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.

3.60.2.2 void VoiceDetectorCalibrate (int durationMs)

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.
-------------------	--

Implements [ILocalVoiceAudio](#).

3.60.3 Property Documentation

3.60.3.1 bool VoiceDetectorCalibrating [get]

True if the `VoiceDetector` is currently calibrating.

3.61 LocalVoiceAudioDummy Class Reference

Dummy [LocalVoiceAudio](#)

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

Public Member Functions

- void [VoiceDetectorCalibrate](#) (int durationMs)
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

3.61.1 Detailed Description

Dummy [LocalVoiceAudio](#)

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

3.61.2 Member Function Documentation

3.61.2.1 void VoiceDetectorCalibrate (int *durationMs*)

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).
-------------------	--

Implements [ILocalVoiceAudio](#).

3.61.3 Member Data Documentation

3.61.3.1 LocalVoiceAudioDummy Dummy = new LocalVoiceAudioDummy() [static]

A Dummy [LocalVoiceAudio](#) instance.

3.62 LocalVoiceAudioFloat Class Reference

Specialization of [LocalVoiceAudio](#) for float audio

Inherits [LocalVoiceAudio< T >](#).

Additional Inherited Members

3.62.1 Detailed Description

Specialization of [LocalVoiceAudio](#) for float audio

3.63 LocalVoiceAudioShort Class Reference

Specialization of [LocalVoiceAudio](#) for short audio

Inherits [LocalVoiceAudio< T >](#).

Additional Inherited Members

3.63.1 Detailed Description

Specialization of [LocalVoiceAudio](#) for short audio

3.64 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'.

Properties

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

Additional Inherited Members

3.64.1 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>channelId</i>	Transport channel specific to frontend. Set to VoiceClient.ChannelAuto to let frontend automatically assign channel.
<i>encoder</i>	Encoder producing the stream.

Returns

Outgoing stream handler.

3.64.2 Member Function Documentation

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

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

Parameters

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

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

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

Parameters

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

3.64.2.3 void ClearProcessors ()

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

3.64.2.4 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](#).

3.64.2.5 void PushData (T[] *buf*)

Synchronously push data into this stream.

3.64.2.6 void PushDataAsync (T[] *buf*)

Asynchronously push data into this stream.

3.64.3 Property Documentation

3.64.3.1 bool PushDataAsyncReady [get]

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

3.65 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

3.65.1 Detailed Description

Typed re-framing [LocalVoice](#)

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

3.65.2 Property Documentation

3.65.2.1 `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.

3.66 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)

3.67 MicWrapper Class Reference

Inherits [IAudioReader](#)< T >.

Public Member Functions

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

Properties

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

3.68 `ObjectFactory< TType, TInfo >` Interface Template Reference

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

Inherits `IDisposable`.

Inherited by [FactoryPrimitiveArrayPool](#)< T >, and [FactoryReusableArray](#)< T >.

Public Member Functions

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

Properties

- TInfo **Info** [get]

3.68.1 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).

3.69 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.

Inherited by [ImageBufferNativePool< T >](#), and [PrimitiveArrayPool< T >](#).

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.

3.69.1 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).

3.69.2 Constructor & Destructor Documentation

3.69.2.1 **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.

3.69.2.2 **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.

3.69.3 Member Function Documentation

3.69.3.1 **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.

3.69.3.2 **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.
-------------	---

3.69.3.3 void Dispose ()

Free resources assoicated with this [ObjectPool](#)

3.69.3.4 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.
-------------	---------------------------------

3.69.3.5 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.

3.69.3.6 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.
------------	-----------------------------------

3.69.4 Property Documentation

3.69.4.1 TInfo Info [get]

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

3.70 OpusCodec Class Reference

Classes

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

Public Types

- enum **FrameDuration**

3.71 OpusDecoder Class Reference

Inherits IDisposable.

Public Member Functions

- **OpusDecoder** (SamplingRate outputSamplingRateHz, [Channels](#) numChannels)
- float[] **DecodePacketFloat** (byte[] packetData)
- short[] **DecodePacketShort** (byte[] packetData)
- void **Dispose** ()

Properties

- string **Version** [get]
- [Bandwidth](#) **PreviousPacketBandwidth** [get]

3.72 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** ()

Public Attributes

- const int **BitrateMax** = -1

Properties

- SamplingRate **InputSamplingRate** [get]
- [Channels](#) **InputChannels** [get]
- string **Version** [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]
- bool **UseUnconstrainedVBR** [get, set]
- bool **DtxEnabled** [get, set]

3.72.1 Property Documentation

3.72.1.1 Delay EncoderDelay [get], [set]

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

3.73 OpusException Class Reference

Inherits Exception.

Public Member Functions

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

Properties

- OpusStatusCode **StatusCode** [get]

3.74 WebRTCAudioLib.Param Struct Reference

Public Attributes

- const int **REVERSE_STREAM_DELAY_MS** = 1
- const int **AEC** = 10
- const int **AEC_SUPPRESSION_LEVEL** = 11
- const int **AECM** = 20
- const int **AECM_ROUTING_MODE** = 21
- const int **AECM_COMFORT_NOISE** = 22
- const int **HIGH_PASS_FILTER** = 31
- const int **NS** = 41
- const int **NS_LEVEL** = 42
- const int **AGC** = 51
- const int **AGC_MODE** = 52
- const int **AGC_COMPRESSION_GAIN** = 56
- const int **AGC_LIMITER** = 57
- const int **VAD** = 61
- const int **VAD_FRAME_SIZE_MS** = 62
- const int **VAD_LIKEHOOD** = 63

3.75 Recorder.PhotonVoiceCreatedParams Class Reference

Properties

- LocalVoice **Voice** [get, set]
- IAudioDesc **AudioDesc** [get, set]

3.76 PhotonVoiceLagSimulationGui Class Reference

Inherits MonoBehaviour.

Public Member Functions

- void **Start** ()
- void **OnGUI** ()

Public Attributes

- Rect **WindowRect** = new Rect(0, 100, 120, 100)
Positioning rect for window.
- int **WindowId** = 101
Unity GUI Window ID (must be unique or will cause issues).
- bool **Visible** = true
Shows or hides GUI (does not affect settings).

Properties

- PhotonPeer **Peer** [get, set]
The peer currently in use (to set the network simulation).

3.76.1 Member Data Documentation

3.76.1.1 bool Visible = true

Shows or hides GUI (does not affect settings).

3.76.1.2 int WindowId = 101

Unity GUI Window ID (must be unique or will cause issues).

3.76.1.3 Rect WindowRect = new Rect(0, 100, 120, 100)

Positioning rect for window.

3.76.2 Property Documentation

3.76.2.1 PhotonPeer Peer [get], [set]

The peer currently in use (to set the network simulation).

3.77 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

- const string [VoiceRoomNameSuffix](#) = "_voice_"
Suffix for voice room names appended to [PUN](#) room names.
- 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 [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](#) ()
- override void [OnApplicationQuit](#) ()
- override void [OnDestroy](#) ()
- override void [OnVoiceStateChanged](#) ([ClientState](#) fromState, [ClientState](#) toState)

Properties

- static [PhotonVoiceNetwork Instance](#) [get, set]
Singleton instance for [PhotonVoiceNetwork](#)

Additional Inherited Members

3.77.1 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.

3.77.2 Member Function Documentation

3.77.2.1 bool ConnectAndJoinRoom ()

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

Returns

If true, connection command send from client

3.77.2.2 void Disconnect ()

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

3.77.3 Member Data Documentation

3.77.3.1 bool AutoConnectAndJoin = true

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

3.77.3.2 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

3.77.3.3 bool AutoLeaveAndDisconnect = true

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

3.77.3.4 const string VoiceRoomNameSuffix = "_voice_"

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

3.77.4 Property Documentation

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

Singleton instance for [PhotonVoiceNetwork](#)

3.78 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 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, protected set]
If true, this [PhotonVoiceView](#) is setup and ready to be used
- bool [IsSpeaker](#) [get, protected set]
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, protected set]
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

Additional Inherited Members

3.78.1 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.

3.78.2 Member Data Documentation

3.78.2.1 bool AutoCreateRecorderIfNotFound

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

3.78.2.2 bool SetupDebugSpeaker

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

3.78.2.3 bool UsePrimaryRecorder

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

3.78.3 Property Documentation

3.78.3.1 bool IsRecorder [get], [protected set]

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

3.78.3.2 bool IsRecording [get]

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

3.78.3.3 `bool IsSetup` `[get], [protected set]`

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

3.78.3.4 `bool IsSpeaker` `[get], [protected set]`

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

3.78.3.5 `bool IsSpeaking` `[get]`

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

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

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

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

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

3.79 `PrimitiveArrayPool< T >` Class Template Reference

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

Inherits [ObjectPool< TType, TInfo >](#).

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

3.79.1 Detailed Description

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

Template Parameters

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

3.80 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 [ReInit](#) ()
Reinitializes the [Recorder](#) if something has changed that requires this.
- void [VoiceDetectorCalibrate](#) (int durationMs)
Trigger voice detector calibration process. While calibrating, keep silence. [Voice](#) detector sets threshold basing on measured background noise level.

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.
- bool [RequiresInit](#) [get]
Returns true if something has changed in the [Recorder](#) after initialization that won't take effect unless re initialized.
- 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.
 - 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.
 - POpusCodec.Enums.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.

Additional Inherited Members

3.80.1 Detailed Description

Component representing outgoing audio stream in scene.

3.80.2 Member Function Documentation

3.80.2.1 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

3.80.2.2 void ReInit ()

Reinitializes the [Recorder](#) if something has changed that requires this.

3.80.2.3 void VoiceDetectorCalibrate (int durationMs)

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.
-------------------	--

3.80.3 Property Documentation

3.80.3.1 AudioClip AudioClip [get], [set]

Source audio clip.

3.80.3.2 byte AudioGroup [get], [set]

Target interest group that will receive transmitted audio.

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

3.80.3.3 int Bitrate [get], [set]

Outgoing audio stream bitrate.

3.80.3.4 bool DebugEchoMode [get], [set]

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

3.80.3.5 bool Encrypt [get], [set]

If true, voice stream is sent encrypted.

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

Outgoing audio stream encoder delay.

3.80.3.7 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

3.80.3.8 bool IsCurrentlyTransmitting [get]

Returns true if audio stream broadcasts.

3.80.3.9 bool IsInitialized [get]

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

3.80.3.10 AudioUtil.ILevelMeter LevelMeter [get]

Level meter utility.

3.80.3.11 bool LoopAudioClip [get], [set]

Loop playback for audio clip sources.

3.80.3.12 MicType MicrophoneType [get], [set]

Which microphone API to use when the Source is set to Microphone.

3.80.3.13 int PhotonMicrophoneDeviceId [get], [set]

Set or get photon microphone device used for streaming.

3.80.3.14 AudioInEnumerator PhotonMicrophoneEnumerator [static], [get]

Enumerator for the available microphone devices gathered by the [Photon](#) plugin.

3.80.3.15 bool ReliableMode [get], [set]

If true, stream data sent in reliable mode.

3.80.3.16 bool RequiresInit [get]

Returns true if something has changed in the [Recorder](#) after initialization that won't take effect unless re initialized.

3.80.3.17 POpusCodec.Enums.SamplingRate SamplingRate [get], [set]

Outgoing audio stream sampling rate.

3.80.3.18 InputSourceType SourceType [get], [set]

Audio data source.

3.80.3.19 bool TransmitEnabled [get], [set]

If true, audio transmission is enabled.

3.80.3.20 SampleTypeConv TypeConvert [get], [set]

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

3.80.3.21 string UnityMicrophoneDevice [get], [set]

Set or get [Unity](#) microphone device used for streaming.

3.80.3.22 object UserData [get], [set]

Custom user object to be sent in the voice stream info event.

3.80.3.23 bool VoiceDetection [get], [set]

If true, voice detection enabled.

3.80.3.24 int VoiceDetectionDelayMs [get], [set]

Keep detected state during this time after signal level dropped below threshold. Default is 500ms

3.80.3.25 float VoiceDetectionThreshold [get], [set]

[Voice](#) detection threshold (0..1, where 1 is full amplitude).

3.80.3.26 AudioUtil.IVoiceDetector VoiceDetector [get]

Returns voice activity detector for recorder's audio stream.

3.80.3.27 bool VoiceDetectorCalibrating [get]

If true, voice detector calibration is in progress.

3.81 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).

3.81.1 Detailed Description

Information about a remote voice (incoming stream).

3.81.2 Property Documentation

3.81.2.1 `int ChannelId` `[get]`

ID of channel used for transmission.

3.81.2.2 `VoiceInfo Info` `[get]`

Remote voice info.

3.81.2.3 `int PlayerId` `[get]`

Player ID of voice owner.

3.81.2.4 `byte VoiceId` `[get]`

[Voice](#) ID (unique in the room).

3.82 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< float[]> **FloatFrameDecoded**
- Action **RemoteVoiceRemoved**

3.83 RemoteVoiceOptions Struct Reference

Event Actions and other options for a remote voice (incoming stream).

Properties

- Action< byte[] > [OnDecodedFrameByteAction](#) [get, set]
Register a method to be called when new data frame received. Use it to get uncompressed data as byte[].
- Action< float[] > [OnDecodedFrameFloatAction](#) [get, set]
Register a method to be called when new data frame received. Use it to get uncompressed data as float[].
- Action< short[] > [OnDecodedFrameShortAction](#) [get, set]
Register a method to be called when new data frame received. Use it to get uncompressed data as short[].
- Action [OnRemoteVoiceRemoveAction](#) [get, set]
Register a method to be called when the remote voice is removed.
- object [LocalUserObject](#) [get, set]
User object (e.g. audio player) attached to remote voice instance for easy access.
- [IDecoder Decoder](#) [get, set]
Remote voice data decoder. Use to set decoder options or override it with user decoder.

3.83.1 Detailed Description

Event Actions and other options for a remote voice (incoming stream).

3.83.2 Property Documentation

3.83.2.1 [IDecoder Decoder](#) [get], [set]

Remote voice data decoder. Use to set decoder options or override it with user decoder.

3.83.2.2 object [LocalUserObject](#) [get], [set]

User object (e.g. audio player) attached to remote voice instance for easy access.

3.83.2.3 Action<byte[] > [OnDecodedFrameByteAction](#) [get], [set]

Register a method to be called when new data frame received. Use it to get uncompressed data as byte[].

3.83.2.4 Action<float[] > [OnDecodedFrameFloatAction](#) [get], [set]

Register a method to be called when new data frame received. Use it to get uncompressed data as float[].

3.83.2.5 Action<short[] > [OnDecodedFrameShortAction](#) [get], [set]

Register a method to be called when new data frame received. Use it to get uncompressed data as short[].

3.83.2.6 Action [OnRemoteVoiceRemoveAction](#) [get], [set]

Register a method to be called when the remote voice is removed.

3.84 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**

3.84.1 Detailed Description

Sample-rate conversion Audio Processor.

This processor converts the sample-rate of the source stream. Internally, it uses AudioUtil.Resample.

3.84.2 Constructor & Destructor Documentation

3.84.2.1 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.

3.84.3 Member Function Documentation

3.84.3.1 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

Implements [IProcessor< T >](#).

3.85 Speaker Class Reference

Component representing remote audio stream in local scene.

Inherits [VoiceComponent](#).

Public Attributes

- int **PlayDelayMs** = 200

Protected Member Functions

- override void **Awake** ()

Properties

- 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.*

Additional Inherited Members

3.85.1 Detailed Description

Component representing remote audio stream in local scene.

3.85.2 Property Documentation

3.85.2.1 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**.

3.85.2.2 bool IsLinked [get]

Whether or not this **Speaker** has been linked to a remote voice stream.

3.85.2.3 bool IsPlaying [get]

Is the speaker playing right now.

3.85.2.4 int Lag [get]

Smoothed difference between (jittering) stream and (clock-driven) audioOutput.

3.85.2.5 Action<Speaker> OnRemoteVoiceRemoveAction [get], [set]

Register a method to be called when remote voice removed.

3.86 SpeexLib Class Reference

Inherited by **SpeexProcessor**.

Public Member Functions

- static IntPtr **speex_preprocess_state_init** (int frame_size, int sampling_rate)
- static void **speex_preprocess_state_destroy** (IntPtr st)
- static int **speex_preprocess_run** (IntPtr st, short[] x)
- static int **speex_preprocess_ctl** (IntPtr st, int request, IntPtr ptr)
- static IntPtr **speex_echo_state_init** (int frame_size, int filter_length)
- static IntPtr **speex_echo_state_init_mc** (int frame_size, int filter_length, int nb_mic, int nb_speakers)
- static void **speex_echo_state_destroy** (IntPtr st)
- static void **speex_echo_cancellation** (IntPtr st, short[] rec, short[] play, short[] outBuf)
- static void **speex_echo_capture** (IntPtr st, short[] rec, short[] outBuf)
- static void **speex_echo_playback** (IntPtr st, short[] play)
- static void **speex_echo_state_reset** (IntPtr st)
- static int **speex_echo_ctl** (IntPtr st, int request, IntPtr ptr)

Static Public Member Functions

- static int **speex_preprocess_ctl** (IntPtr st, int request, ref int value)
- static int **speex_preprocess_ctl** (IntPtr st, int request, ref float value)
- static int **speex_echo_ctl** (IntPtr st, int request, ref int value)
- static int **speex_echo_ctl** (IntPtr st, int request, ref float value)

Public Attributes

- const int **SPEEX_PREPROCESS_SET_DENOISE** = 0
- const int **SPEEX_PREPROCESS_GET_DENOISE** = 1
- const int **SPEEX_PREPROCESS_SET_AGC** = 2
- const int **SPEEX_PREPROCESS_GET_AGC** = 3
- const int **SPEEX_PREPROCESS_SET_VAD** = 4
- const int **SPEEX_PREPROCESS_GET_VAD** = 5
- const int **SPEEX_PREPROCESS_SET_AGC_LEVEL** = 6
- const int **SPEEX_PREPROCESS_GET_AGC_LEVEL** = 7
- const int **SPEEX_PREPROCESS_SET_DEREVERB** = 8
- const int **SPEEX_PREPROCESS_GET_DEREVERB** = 9
- const int **SPEEX_PREPROCESS_SET_DEREVERB_LEVEL** = 10
- const int **SPEEX_PREPROCESS_GET_DEREVERB_LEVEL** = 11
- const int **SPEEX_PREPROCESS_SET_DEREVERB_DECAY** = 12
- const int **SPEEX_PREPROCESS_GET_DEREVERB_DECAY** = 13
- const int **SPEEX_PREPROCESS_SET_PROB_START** = 14
- const int **SPEEX_PREPROCESS_GET_PROB_START** = 15
- const int **SPEEX_PREPROCESS_SET_PROB_CONTINUE** = 16
- const int **SPEEX_PREPROCESS_GET_PROB_CONTINUE** = 17
- const int **SPEEX_PREPROCESS_SET_NOISE_SUPPRESS** = 18
- const int **SPEEX_PREPROCESS_GET_NOISE_SUPPRESS** = 19
- const int **SPEEX_PREPROCESS_SET_ECHO_SUPPRESS** = 20
- const int **SPEEX_PREPROCESS_GET_ECHO_SUPPRESS** = 21
- const int **SPEEX_PREPROCESS_SET_ECHO_SUPPRESS_ACTIVE** = 22
- const int **SPEEX_PREPROCESS_GET_ECHO_SUPPRESS_ACTIVE** = 23
- const int **SPEEX_PREPROCESS_SET_ECHO_STATE** = 24
- const int **SPEEX_PREPROCESS_GET_ECHO_STATE** = 25
- const int **SPEEX_PREPROCESS_SET_AGC_INCREMENT** = 26
- const int **SPEEX_PREPROCESS_GET_AGC_INCREMENT** = 27
- const int **SPEEX_PREPROCESS_SET_AGC_DECREMENT** = 28
- const int **SPEEX_PREPROCESS_GET_AGC_DECREMENT** = 29

- const int [SPEEX_PREPROCESS_SET_AGC_MAX_GAIN](#) = 30
- const int [SPEEX_PREPROCESS_GET_AGC_MAX_GAIN](#) = 31
- const int [SPEEX_PREPROCESS_GET_AGC_LOUDNESS](#) = 33
- const int [SPEEX_PREPROCESS_GET_AGC_GAIN](#) = 35
- const int [SPEEX_PREPROCESS_GET_PSD_SIZE](#) = 37
- const int [SPEEX_PREPROCESS_GET_PSD](#) = 39
- const int [SPEEX_PREPROCESS_GET_NOISE_PSD_SIZE](#) = 41
- const int [SPEEX_PREPROCESS_GET_NOISE_PSD](#) = 43
- const int [SPEEX_PREPROCESS_GET_PROB](#) = 45
- const int [SPEEX_PREPROCESS_SET_AGC_TARGET](#) = 46
- const int [SPEEX_PREPROCESS_GET_AGC_TARGET](#) = 47
- const int [SPEEX_ECHO_GET_FRAME_SIZE](#) = 3
- const int [SPEEX_ECHO_SET_SAMPLING_RATE](#) = 24
- const int [SPEEX_ECHO_GET_SAMPLING_RATE](#) = 25
- const int [SPEEX_ECHO_GET_IMPULSE_RESPONSE_SIZE](#) = 27
- const int [SPEEX_ECHO_GET_IMPULSE_RESPONSE](#) = 29

3.86.1 Member Data Documentation

3.86.1.1 const int [SPEEX_ECHO_GET_FRAME_SIZE](#) = 3

Obtain frame size used by the AEC

3.86.1.2 const int [SPEEX_ECHO_GET_IMPULSE_RESPONSE](#) = 29

Get impulse response (int32[])

3.86.1.3 const int [SPEEX_ECHO_GET_IMPULSE_RESPONSE_SIZE](#) = 27

Get size of impulse response (int32)

3.86.1.4 const int [SPEEX_ECHO_GET_SAMPLING_RATE](#) = 25

Get sampling rate

3.86.1.5 const int [SPEEX_ECHO_SET_SAMPLING_RATE](#) = 24

Set sampling rate

3.86.1.6 const int [SPEEX_PREPROCESS_GET_AGC](#) = 3

Get preprocessor Automatic Gain Control state

3.86.1.7 const int [SPEEX_PREPROCESS_GET_AGC_DECREMENT](#) = 29

Get maximal gain decrease in dB/second (int32)

3.86.1.8 const int [SPEEX_PREPROCESS_GET_AGC_GAIN](#) = 35

Get current gain (int32 percent)

3.86.1.9 `const int SPEEX_PREPROCESS_GET_AGC_INCREMENT = 27`

Get maximal gain increase in dB/second (int32)

3.86.1.10 `const int SPEEX_PREPROCESS_GET_AGC_LEVEL = 7`

Get preprocessor Automatic Gain Control level (float)

3.86.1.11 `const int SPEEX_PREPROCESS_GET_AGC_LOUDNESS = 33`

Get loudness

3.86.1.12 `const int SPEEX_PREPROCESS_GET_AGC_MAX_GAIN = 31`

Get maximal gain in dB (int32)

3.86.1.13 `const int SPEEX_PREPROCESS_GET_AGC_TARGET = 47`

Get preprocessor Automatic Gain Control level (int32)

3.86.1.14 `const int SPEEX_PREPROCESS_GET_DENOISE = 1`

Get preprocessor denoiser state

3.86.1.15 `const int SPEEX_PREPROCESS_GET_DEREVERB = 9`

Get preprocessor dereverb state

3.86.1.16 `const int SPEEX_PREPROCESS_GET_DEREVERB_DECAY = 13`

Get preprocessor dereverb decay

3.86.1.17 `const int SPEEX_PREPROCESS_GET_DEREVERB_LEVEL = 11`

Get preprocessor dereverb level

3.86.1.18 `const int SPEEX_PREPROCESS_GET_ECHO_STATE = 25`

Get the corresponding echo canceller state

3.86.1.19 `const int SPEEX_PREPROCESS_GET_ECHO_SUPPRESS = 21`

Get maximum attenuation of the residual echo in dB (negative number)

3.86.1.20 `const int SPEEX_PREPROCESS_GET_ECHO_SUPPRESS_ACTIVE = 23`

Get maximum attenuation of the residual echo in dB when near end is active (negative number)

3.86.1.21 `const int SPEEX_PREPROCESS_GET_NOISE_PSD = 43`

Get noise estimate (int32[] of squared values)

3.86.1.22 `const int SPEEX_PREPROCESS_GET_NOISE_PSD_SIZE = 41`

Get spectrum size for noise estimate (int32)

3.86.1.23 `const int SPEEX_PREPROCESS_GET_NOISE_SUPPRESS = 19`

Get maximum attenuation of the noise in dB (negative number)

3.86.1.24 `const int SPEEX_PREPROCESS_GET_PROB = 45`

Get speech probability in last frame (int32).

3.86.1.25 `const int SPEEX_PREPROCESS_GET_PROB_CONTINUE = 17`

Get probability required for the VAD to stay in the voice state (integer percent)

3.86.1.26 `const int SPEEX_PREPROCESS_GET_PROB_START = 15`

Get probability required for the VAD to go from silence to voice

3.86.1.27 `const int SPEEX_PREPROCESS_GET_PSD = 39`

Get power spectrum (int32[] of squared values)

3.86.1.28 `const int SPEEX_PREPROCESS_GET_PSD_SIZE = 37`

Get spectrum size for power spectrum (int32)

3.86.1.29 `const int SPEEX_PREPROCESS_GET_VAD = 5`

Get preprocessor [Voice](#) Activity Detection state

3.86.1.30 `const int SPEEX_PREPROCESS_SET_AGC = 2`

Set preprocessor Automatic Gain Control state

3.86.1.31 `const int SPEEX_PREPROCESS_SET_AGC_DECREMENT = 28`

Set maximal gain decrease in dB/second (int32)

3.86.1.32 `const int SPEEX_PREPROCESS_SET_AGC_INCREMENT = 26`

Set maximal gain increase in dB/second (int32)

3.86.1.33 `const int SPEEX_PREPROCESS_SET_AGC_LEVEL = 6`

Set preprocessor Automatic Gain Control level (float)

3.86.1.34 `const int SPEEX_PREPROCESS_SET_AGC_MAX_GAIN = 30`

Set maximal gain in dB (int32)

3.86.1.35 `const int SPEEX_PREPROCESS_SET_AGC_TARGET = 46`

Set preprocessor Automatic Gain Control level (int32)

3.86.1.36 `const int SPEEX_PREPROCESS_SET_DENOISE = 0`

Set preprocessor denoiser state

3.86.1.37 `const int SPEEX_PREPROCESS_SET_DEREVERB = 8`

Set preprocessor dereverb state

3.86.1.38 `const int SPEEX_PREPROCESS_SET_DEREVERB_DECAY = 12`

Set preprocessor dereverb decay

3.86.1.39 `const int SPEEX_PREPROCESS_SET_DEREVERB_LEVEL = 10`

Set preprocessor dereverb level

3.86.1.40 `const int SPEEX_PREPROCESS_SET_ECHO_STATE = 24`

Set the corresponding echo canceller state so that residual echo suppression can be performed (NULL for no residual echo suppression)

3.86.1.41 `const int SPEEX_PREPROCESS_SET_ECHO_SUPPRESS = 20`

Set maximum attenuation of the residual echo in dB (negative number)

3.86.1.42 `const int SPEEX_PREPROCESS_SET_ECHO_SUPPRESS_ACTIVE = 22`

Set maximum attenuation of the residual echo in dB when near end is active (negative number)

3.86.1.43 `const int SPEEX_PREPROCESS_SET_NOISE_SUPPRESS = 18`

Set maximum attenuation of the noise in dB (negative number)

3.86.1.44 `const int SPEEX_PREPROCESS_SET_PROB_CONTINUE = 16`

Set probability required for the VAD to stay in the voice state (integer percent)

3.86.1.45 `const int SPEEX_PREPROCESS_SET_PROB_START = 14`

Set probability required for the VAD to go from silence to voice

3.86.1.46 `const int SPEEX_PREPROCESS_SET_VAD = 4`

Set preprocessor [Voice](#) Activity Detection state

3.87 SpeexProcessor Class Reference

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

Classes

- struct [AECLatencyResultType](#)

Public Member Functions

- void **ResetAEC** ()
- void **AECLatecnyDetectCaliberate** ()
- **SpeexProcessor** ([ILogger](#) logger, Func< long > clockMs, int frameSize, int samplingRate, int channels, int playSamplingRate, int playChannels, int playBufSize)
- void **InitAEC** ()
- short[] **Process** (short[] buf)
- void **OnAudioOutFrame** (float[] data, int outChannels)
- void **PrintInfo** ()
- void **Dispose** ()

Properties

- bool **AEC** [get, set]
- int **AECFilterLengthMs** [get, set]
- int **AECPlaybackDelayMs** [get, set]
- int **AECCurrentPlayDelayFrames** [get]
- bool **AECLatencyDetect** [get, set]
- [AECLatencyResultType](#) **AECLatencyResult** [get]
- bool **Denoise** [get, set]
- bool **AGC** [get, set]
- float **AGCLevel** [get, set]

Additional Inherited Members

3.88 TestTone Class Reference

Inherits [MonoBehaviour](#).

3.89 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=441000, 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]

3.89.1 Detailed Description

[IAudioPusher](#) that provides a constant tone signal.

3.89.2 Constructor & Destructor Documentation

3.89.2.1 [ToneAudioPusher](#) (int *frequency* = 440, int *bufSizeMs* = 100, int *samplingRate* = 441000, 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.

3.89.3 Member Function Documentation

3.89.3.1 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 >](#).

3.90 [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=441000, 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.

3.90.1 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.

3.90.2 Constructor & Destructor Documentation

3.90.2.1 [ToneAudioReader](#) (Func< double > *clockSec* = null, double *frequency* = 440, int *samplingRate* = 441000, 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.

3.90.3 Member Function Documentation

3.90.3.1 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 >](#).

3.90.4 Property Documentation

3.90.4.1 int Channels [get]

Number of channels in the audio signal.

3.90.4.2 string Error [get]

If not null, audio object is in invalid state.

3.90.4.3 int SamplingRate [get]

Sampling rate of the audio signal (in Hz).

3.91 ToneAudioReader Class Reference

Inherits [IAudioReader< T >](#).

Public Member Functions

- void **Dispose** ()
- bool **Read** (float[] buf)

Properties

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

3.92 UnityAndroidAudioInAEC Class Reference

Inherits [IAudioPusher< T >](#).

Public Member Functions

- **UnityAndroidAudioInAEC** ([Voice.ILogger](#) logger)
- void **SetCallback** (Action< short[]> callback, [ObjectFactory](#)< short[], int > bufferFactory)
- void **Dispose** ()

Properties

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

3.93 UnityAudioOut Class Reference

Inherits [ISyncAudioOut](#).

Public Member Functions

- **UnityAudioOut** (AudioSource audioSource)
- void **Start** (int frequency, int channels, int frameSamples, int playDelayMs)
- void **Service** ()
- void **Push** (float[] frame)
- void **Stop** ()
- void **Pause** ()
- void **UnPause** ()

Public Attributes

- const int **FRAME_POOL_CAPACITY** = 50

Properties

- int **Lag** [get]
- int **PlaySamplePos** [get, set]
- bool **IsPlaying** [get]

3.94 UnsupportedCodecException Class Reference

Exception thrown if an unsupported codec is encountered.

Inherits Exception.

Public Member Functions

- [UnsupportedCodecException](#) (Codec codec, LocalVoice voice)
Create a new [UnsupportedCodecException](#).

3.94.1 Detailed Description

Exception thrown if an unsupported codec is encountered.

PhotonVoice currently only supports one Codec, [Codec.AudioOpus](#).

3.94.2 Constructor & Destructor Documentation

3.94.2.1 UnsupportedCodecException (Codec codec, LocalVoice voice)

Create a new [UnsupportedCodecException](#).

Parameters

<i>codec</i>	The codec actually encountered.
<i>voice</i>	The LocalVoice (outgoing stream) involved.

3.95 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](#).

3.95.1 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.

3.95.2 Constructor & Destructor Documentation

3.95.2.1 [UnsupportedSampleTypeException](#) (Type t)

Create a new [UnsupportedSampleTypeException](#).

Parameters

<i>t</i>	The sample type actually encountered.
----------	---------------------------------------

3.96 OpusCodec.Util Class Reference

3.97 VoiceClient Class Reference

Base class for [Voice](#) clients implamantations

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 [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=ChannelAuto, [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=ChannelAuto, [IEncoderDataFlow](#)< T > 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, int channelId=ChannelAuto, [IEncoder](#) encoder=null)
Creates outgoing audio stream. Adds audio specific features (e.g. resampling, level meter) to processing pipeline and to returning stream handler.
- [Voice.LocalVoice](#) [CreateLocalVoiceAudioFromSource](#) ([Voice.VoiceInfo](#) voiceInfo, [Voice.IAudioDesc](#) source, bool forceShort=false, int channelId=ChannelAuto, [IEncoder](#) encoder=null)

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** ()

Public Attributes

- const int **ChannelAuto** = -1

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 voiceId, [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.
- IEnumerable< object > [RemoteVoiceLocalUserObjects](#) [get]
Iterates through all local objects set by user in remote voices.

3.97.1 Detailed Description

Base class for [Voice](#) clients implamantations

3.97.2 Member Function Documentation

3.97.2.1 [LocalVoice](#) CreateLocalVoice ([VoiceInfo](#) voiceInfo, int channelId = ChannelAuto, [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 frontend. Set to <code>VoiceClient.ChannelAuto</code> to let frontend automatically assign channel.
<i>encoder</i>	Encoder producing the stream.

Returns

Outgoing stream handler.

3.97.2.2 `LocalVoiceAudio<T> CreateLocalVoiceAudio< T > (VoiceInfo voiceInfo, int channelId = ChannelAuto, IEncoder encoder = null)`

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 frontend. Set to <code>VoiceClient.ChannelAuto</code> to let frontend automatically assign channel.
<i>encoder</i>	Audio encoder. Set to null to use default Opus encoder.

Returns

Outgoing stream handler.

`voiceInfo.sourceSamplingRate` and `voiceInfo.SamplingRate` may do not match. Automatic resampling will occur in this case.

3.97.2.3 `Voice.LocalVoice CreateLocalVoiceAudioFromSource (Voice.VoiceInfo voiceInfo, Voice.IAudioDesc source, bool forceShort = false, int channelId = ChannelAuto, IEncoder encoder = null)`

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>forceShort</i>	For audio sources producing buffers of 'float' type, creates stream of 'short' type and adds converter.
<i>channelId</i>	Transport channel specific to frontend. Set to <code>VoiceClient.ChannelAuto</code> to let frontend automatically assign channel.

<i>encoder</i>	Audio encoder. Set to null to use default Opus encoder.
----------------	---

Returns

Outgoing stream handler.

voiceInfo.sourceSamplingRate and voiceInfo.SamplingRate may do not match. Automatic resampling will occur in this case.

3.97.2.4 LocalVoiceFramed<T> CreateLocalVoiceFramed< T > (VoiceInfo voiceInfo, int frameSize, int channelId = ChannelAuto, IEncoderDataFlow< T > 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>channelId</i>	Transport channel specific to frontend. Set to VoiceClient.ChannelAuto to let frontend automatically assign channel.
<i>encoder</i>	Encoder compressing data stream in pipeline.

Returns

Outgoing stream handler.

3.97.2.5 IEnumerable<LocalVoice> LocalVoicesInChannel (int channelId)

Iterates through copy of all local voices list of given channel.

3.97.2.6 delegate void RemoteVoiceInfoDelegate (int channelId, int playerId, byte voiceId, VoiceInfo voiceInfo, ref RemoteVoiceOptions options)

Remote voice info event delegate.

3.97.2.7 void RemoveLocalVoice (LocalVoice voice)

Removes local voice (outgoing data stream).

Parameters

<i>voice</i>	Handler of outgoing stream to be removed.
--------------	---

3.97.2.8 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).

3.97.3 Property Documentation

3.97.3.1 `int DebugLostPercent` `[get]`, `[set]`

Lost frames simulation ratio.

3.97.3.2 `int FramesLost` `[get]`, `[set]`

Lost frames counter.

3.97.3.3 `int FramesReceived` `[get]`

Received frames counter.

3.97.3.4 `int FramesSent` `[get]`

Sent frames counter.

3.97.3.5 `int FramesSentBytes` `[get]`

Sent frames bytes counter.

3.97.3.6 `IEnumerable<LocalVoice> LocalVoices` `[get]`

Iterates through copy of all local voices list.

3.97.3.7 `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);

3.97.3.8 `IEnumerable<RemoteVoiceInfo> RemoteVoiceInfos` `[get]`

Iterates through all remote voices infos.

3.97.3.9 `IEnumerable<object> RemoteVoiceLocalUserObjects` `[get]`

Iterates through all local objects set by user in remote voices.

3.97.3.10 `int RoundTripTime` `[get]`

Average time required voice packet to return to sender.

3.97.3.11 `int RoundTripTimeVariance` `[get]`

Average round trip time variation.

3.97.3.12 `bool SuppressInfoDuplicateWarning` `[get]`, `[set]`

Do not log warning when duplicate info received.

3.98 VoiceComponent Class Reference

Inherits MonoBehaviour, and [ILoggable](#).

Inherited by [PhotonVoiceView](#), [Recorder](#), [Speaker](#), and [WebRtcAudioDsp](#).

Protected Member Functions

- virtual void **Awake** ()

Protected Attributes

- DebugLevel **LogLevel** = DebugLevel.ERROR

Properties

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

3.99 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](#)

Public Attributes

- AppSettings [Settings](#)
Settings to be used by this voice connection
- [Recorder](#) **PrimaryRecorder**
Main [Recorder](#) to be used for transmission by default
- Func< int, byte, object, [Speaker](#) > **SpeakerFactory**
Special factory to link [Speaker](#) components with incoming remote audio streams

Protected Member Functions

- override void **Awake** ()
- virtual void **Update** ()
- override void **OnDestroy** ()
- virtual void **OnVoiceStateChanged** ([ClientState](#) fromState, [ClientState](#) toState)
- override void **OnApplicationQuit** ()
- 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 [LoadBalancingFrontend](#) [Client](#) [get]
Returns underlying [Photon](#) LoadBalancing client.
- [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

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

3.99.1 Detailed Description

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

3.99.2 Member Function Documentation

3.99.2.1 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

3.99.3 Member Data Documentation

3.99.3.1 Recorder PrimaryRecorder

Main [Recorder](#) to be used for transmission by default

3.99.3.2 AppSettings Settings

Settings to be used by this voice connection

3.99.3.3 Func<int, byte, object, Speaker> SpeakerFactory

Special factory to link [Speaker](#) components with incoming remote audio streams

3.99.4 Property Documentation

3.99.4.1 new LoadBalancingFrontend Client [get]

Returns underlying [Photon](#) LoadBalancing client.

3.99.4.2 ClientState ClientState [get]

Returns [Photon Voice](#) client state.

3.99.4.3 float FramesLostPercent [get]

Percentage of lost frames.

3.99.4.4 float FramesLostPerSecond [get]

Number of frames lost per second.

3.99.4.5 float FramesReceivedPerSecond [get]

Number of frames received per second.

3.99.4.6 VoiceLogger Logger [get], [protected set]

[Logger](#) used by this component

3.99.4.7 DebugLevel LogLevel [get], [set]

Log level for this component

3.99.4.8 GameObject SpeakerPrefab [get], [set]

Prefab that contains [Speaker](#) component to be instantiated when receiving a new remote audio source info

3.99.4.9 VoiceClient VoiceClient [get]

Returns underlying [Photon Voice](#) client.

3.99.5 Event Documentation

3.99.5.1 Action<RemoteVoiceLink> RemoteVoiceAdded

Fires when a remote voice stream is added

3.99.5.2 Action<Speaker> SpeakerLinked

Fires when a speaker has been linked to a remote audio stream

3.100 AudioUtil.VoiceDetector< T > Class Template Reference

Simple voice activity detector triggered by signal level.

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

Inherited by [AudioUtil.VoiceDetectorFloat](#), and [AudioUtil.VoiceDetectorShort](#).

Public Member Functions

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

Protected Attributes

- 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.

3.100.1 Detailed Description

Simple voice activity detector triggered by signal level.

3.100.2 Member Function Documentation

3.100.2.1 `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

Implements [IProcessor< T >](#).

3.100.3 Property Documentation

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

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

3.100.3.2 `bool Detected [get],[protected set]`

If true, voice detected.

3.100.3.3 `DateTime DetectedTime [get]`

Last time when switched to detected state.

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

If true, voice detection enabled.

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

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

3.100.4 Event Documentation

3.100.4.1 `Action OnDetected`

Called when switched to detected state.

3.101 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 [VoiceDetectorCalibrate](#) (int durationMs)
Start calibration.
- T[] [Process](#) (T[] buf)
Process a frame of audio data.
- void **Dispose** ()

Protected Attributes

- int **voiceDetectorCalibrateCount**

Properties

- bool **VoiceDetectorCalibrating** [get]

3.101.1 Detailed Description

Calibration Utility for [Voice](#) Detector

Using this audio processor, you can calibrate the [IVoiceDetector.Threshold](#).

3.101.2 Constructor & Destructor Documentation

3.101.2.1 [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.

3.101.3 Member Function Documentation

3.101.3.1 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

Implements [IProcessor< T >](#).

3.101.3.2 void VoiceDetectorCalibrate (int *durationMs*)

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.

3.102 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

3.102.1 Detailed Description

Dummy [VoiceDetector](#) that doesn't actually do anything.

3.103 AudioUtil.VoiceDetectorFloat Class Reference

[VoiceDetector](#) specialization for float audio.

Inherits [AudioUtil.VoiceDetector< T >](#).

Public Member Functions

- [VoiceDetectorFloat](#) (int samplingRate, int numChannels)
Create a new [VoiceDetectorFloat](#) instance.
- override float[] **Process** (float[] buffer)

Additional Inherited Members

3.103.1 Detailed Description

[VoiceDetector](#) specialization for float audio.

3.103.2 Constructor & Destructor Documentation

3.103.2.1 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.

3.104 AudioUtil.VoiceDetectorShort Class Reference

[VoiceDetector](#) specialization for float audio.

Inherits [AudioUtil.VoiceDetector< T >](#).

Public Member Functions

- [VoiceDetectorShort](#) (int samplingRate, int numChannels)
Create a new [VoiceDetectorFloat](#) instance
- override short[] **Process** (short[] buffer)

Additional Inherited Members

3.104.1 Detailed Description

[VoiceDetector](#) specialization for float audio.

3.104.2 Constructor & Destructor Documentation

3.104.2.1 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.

3.105 VoiceEventCode Class Reference

PhotonVoice communication uses a single type of event, but differentiates transmission Channels by encoding a channelId into [VoiceEventCode](#).

Static Public Member Functions

- static byte [GetCode](#) (int channelId)
Get the event code for the given channel ID.
- static bool [TryGetChannelID](#) (byte evCode, int maxChannels, out byte channelId)
Try to get the channel ID for the given event code.

Public Attributes

- const byte [Code0](#) = 201
Start of voice event codes range.

3.105.1 Detailed Description

PhotonVoice communication uses a single type of event, but differentiates transmission Channels by encoding a channelId into [VoiceEventCode](#).

Transmission Channels are not for selective forwarding: use AudioGroups for that. Instead, they are to differentiate opus audio from (future) other codecs or media.

For this purpose, a range of event codes of length LoadBalancingPeer.ChannelCount, starting from Code0, is used.

3.105.2 Member Function Documentation

3.105.2.1 static byte GetCode (int channelId) [static]

Get the event code for the given channel ID.

Parameters

<i>channelID</i>	Channel ID to get event code for.
------------------	-----------------------------------

Returns

The corresponding event code.

3.105.2.2 static bool TryGetChannelID (byte evCode, int maxChannels, out byte channelId) [static]

Try to get the channel ID for the given event code.

Parameters

<i>evCode</i>	Event code to find Channel ID from.
<i>maxChannels</i>	Maximum Channel ID in use.
<i>channelID</i>	(output) Channel ID found.

Returns

True if a valid channel ID could be recovered from evCode, false otherwise.

3.105.3 Member Data Documentation

3.105.3.1 const byte Code0 = 201

Start of voice event codes range.

Change if it conflicts with other event codes used in the same [Photon](#) room.

3.106 VoicelInfo Struct Reference

Describes stream properties.

Public Member Functions

- override string **ToString** ()

Static Public Member Functions

- static [VoicelInfo CreateAudioOpus](#) (POpusCodec.Enums.SamplingRate samplingRate, int sourceSampling↔Rate, int channels, OpusCodec.FrameDuration frameDurationUs, int bitrate, 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 [SourceSamplingRate](#) [get, set]
Source audio sampling rate (to be resampled to SamplingRate; 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).
- 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) size in samples.
- int [Width](#) [get, set]
Video width (optional).
- int [Height](#) [get, set]
Video height (optional)

3.106.1 Detailed Description

Describes stream properties.

3.106.2 Member Function Documentation

- 3.106.2.1 static [VoicelInfo CreateAudioOpus](#) (POpusCodec.Enums.SamplingRate *samplingRate*, int *sourceSamplingRate*, 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>source↔SamplingRate</i>	Source audio sampling rate (to be resampled to samplingRate; in Hz).
<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.

3.106.3 Property Documentation

3.106.3.1 int Bitrate [get], [set]

Target bitrate (in bits/second).

3.106.3.2 int Channels [get], [set]

Number of channels.

3.106.3.3 int FrameDurationSamples [get]

Uncompressed frame (data packet) size in samples.

3.106.3.4 int FrameDurationUs [get], [set]

Uncompressed frame (audio packet) size in microseconds.

3.106.3.5 int FrameSize [get]

Uncompressed frame (data packet) size in samples.

3.106.3.6 int Height [get], [set]

Video height (optional)

3.106.3.7 int SamplingRate [get], [set]

Audio sampling rate (frequency, in Hz).

3.106.3.8 int SourceSamplingRate [get], [set]

Source audio sampling rate (to be resampled to SamplingRate; in Hz).

3.106.3.9 object UserData [get], [set]

Optional user data. Should be serializable by [Photon](#).

3.106.3.10 int Width [get], [set]

Video width (optional).

3.107 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*)
Start calibration
- T[] [Process](#) (T[] *buf*)
Process a frame of audio data.
- void **Dispose** ()

Properties

- [ILevelMeter Level](#) [get]
The [LevelMeter](#) in use.
- [IVoiceDetector Detector](#) [get]
The [VoiceDetector](#) in use

3.107.1 Detailed Description

Utility Audio Processor [Voice](#) Detection Calibration.

Encapsulates level meter, voice detector and voice detector calibrator in single instance.

3.107.2 Constructor & Destructor Documentation

3.107.2.1 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.

3.107.3 Member Function Documentation

3.107.3.1 void Calibrate (int *durationMs*)

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.

3.107.3.2 `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

Implements [IProcessor< T >](#).

3.107.4 Property Documentation

3.107.4.1 IVoiceDetector Detector [get]

The [VoiceDetector](#) in use

3.107.4.2 ILevelMeter Level [get]

The [LevelMeter](#) in use.

3.108 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]

3.109 WebRtcAudioDsp Class Reference

Inherits [VoiceComponent](#).

Protected Member Functions

- override void **Awake** ()

Properties

- bool **AEC** [get, set]
- bool **AECMobile** [get, set]
- int **ReverseStreamDelayMs** [get, set]
- bool **NoiseSuppression** [get, set]
- bool **HighPass** [get, set]
- bool **Bypass** [get, set]
- bool **AGC** [get, set]
- bool **VAD** [get, set]

Additional Inherited Members

3.110 WebRTCAudioLib Class Reference

Inherited by [WebRTCAudioProcessor](#).

Classes

- struct [ConfigParam](#)
- struct [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_set_config_param** (IntPtr proc, int param, int v)
- 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)

3.111 WebRTCAudioProcessor Class Reference

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

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** ()

Properties

- int **AECStreamDelayMs** [set]
- bool **AEC** [set]
- bool **AECMobile** [set]
- int **AECMRoutingMode** [set]
- bool **AECMComfortNoise** [set]
- bool **HighPass** [set]
- bool **NoiseSuppression** [set]
- bool **AGC** [set]
- bool **VAD** [set]
- bool **Bypass** [set]

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