



Agora Windows SDK Reference

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Required Libraries

- Agora Audio SDK requires Visual C++ 2008 x86 runtime libraries.
- Add the AgoraAudioSDK/include directory to the INCLUDE directories of your project.
- Add the 'AgoraAudioSDK/lib' directory to the LIB directories of your project and make sure mediasdk.lib is linked with your project.
- Copy .dlls under AgoraAudioSDK/dll to the directory where your executable file is located.

AgoraAudio Methods

Create Agora Audio Object

```
AgoraAudio = createAgoraAudioInstance(IAgoraAudioEventHandler* pHandler);
```

This statement initializes the AgoraAudio class.

| Name | Description |
|--------------|-------------------------------------|
| pHandler | IAgoraAudioEventHandler (See below) |
| Return Value | IAgoraAudio object |

Release Agora Audio Object

This method destroys the AgoraAudio object.

void release()

Join Channel

This method lets users join a channel. Think of it as joining a chat room, except that it is a multi-party phone call. This method is asynchronous, so it can be called on the main UI thread.

void joinChannel(const char* vendorKey, const char* channelName, const char* info, unsigned int uid)

| Name | Description |
|-------------|---|
| vendorKey | Account credentials issued by Agora Voice to app developer, i.e., a user license. |
| channelName | Channel name. Arbitrary descriptor like "game1" or "call2". |

| | |
|------|---|
| info | Optional. Whatever the additional information the programmer wants to add. |
| uid | Optional. User id. If you do not set this parameter, the object instance will automatically generate one. |

Leave Channel

Leave channel, meaning hang up or exit call. Returns 0 if the client successfully left the channel.

int leave()

Set Parameters

void setParameters(const char* parameters)

The following are available parameters used for the Agora Audio engine. The parameters must be in JSON format when specifying new parameters to set. Instead of being called directly by the app, it is usually called by the helper class AgoraAudioParameters.

| Name | Description |
|------------|---|
| parameters | Parameters in JSON format: mute mutePeers speakerOn speakerVolume micVolume enableVolumeReport volumeSmoothFactor logFilter |

Get Parameters

int getParameters(const char* parameters, char* buffer, size_t* length)

Retrieve the current parameters settings.

| Name | Description |
|------------|---------------------------------------|
| parameters | Indicate which parameters to retrieve |
| buffer | String containing values |

| | |
|--------|------------------|
| length | Length of buffer |
|--------|------------------|

Get Call ID

const char* getCallId()

Retrieve current call ID.

Rate For The Call

int rate(const char* callId, int rating)

This method lets the user rate the call.

| Name | Description |
|--------------|---|
| callId | Call ID got from getCallId method |
| Rating | Rating for the call, range from 1(min) to 10(max) |
| Return Value | 0: successful method call -1: method call failed |

Complain About The Call

int complain(const char* callId)

This method lets the user complain about the call.

| Name | Description |
|--------------|---|
| callId | Call ID got from getCallId method |
| Return Value | 0: successful method call -1: method call failed |

Start Echo Test

int StartEchoTest(const char* key)

This method lets the user launch an echo test, paired with StopEchoTest. In an echo test, the user says something and then it's played back so that call connectivity and quality can be tested. (NOTE: echo test should be stopped before starting a call)

| Name | Description |
|------|--|
| key | Account credentials issued by Agora Voice to app developer, i.e., a user |

| | |
|--------------|---|
| Return Value | 0: successful method call -1: method call failed |
|--------------|---|

Stop Echo Test

int StopEchoTest()

This method stops an echo test.

| Name | Description |
|--------------|---|
| Return Value | 0: successful method call -1: method call failed |

Enable Network Test

int enableNetworkTest(const char* key)

This method launches a test to monitor the real-time user network quality. The network test will stop when the app is put into a background state. (NOTE: Network testing is off by default)

| Name | Description |
|--------------|--|
| key | Account credentials issued by Agora Voice to app developer, i.e., a user |
| Return Value | 0: successful method call -1: method call failed |

Disable Network Test

int disableNetworkTest()

This method disables the network test.

| Name | Description |
|--------------|---|
| Return Value | 0: successful method call -1: method call failed |

Make Call Quality Report URL

int makeQualityReportUrl(const char* vendorKey, const char* channel, uid_t listenerUid, uid_t speakerUid, int format, char* buffer, size_t* length)

This method lets the user retrieve the call quality report URL.

| Name | Description |
|--------------|---|
| vendorKey | Account credentials issued by Agora Voice to app developer, i.e., a user |
| channel | The channel number specified in joinChannel method. |
| listenerUid | The user ID - This could be the uid specified in joinChannel method. If it is not set, the SDK object instance will allocate one and the user can retrieve this ID from onJoinSuccess callback (see below). |
| speakerUid | The speaker ID - The user can retrieve it from the onUserJoined callback (see below). If speakerUid is not set, the method call will request a report all speakers in the group call. |
| format | Report format type 0: JSON 1: HTML |
| buffer | User specified input buffer to store the report URL |
| length | Input buffer length |
| Return Value | 0: successful method call -1: method call failed |

AgoraAudioParameters Methods

Mute

void mute(bool mute)

Turns off the microphone.

| Name | Description |
|------|--|
| mute | True - turns microphone off False - turns microphone on |

Mute All Speakers

void mutePeers(bool mute);

Turns off both the speaker and earpiece.

| Name | Description |
|------|--|
| mute | True - turns off all audio output False - turns on all audio output |

Mute Specific User

void mutePeer(bool mute, unsigned int uid);

Turn off audio for a specific caller.

| Name | Description |
|------|--|
| mute | True - muted, False - unmuted (i.e., caller's microphone turned back on) |
| Uid | ID of user to mute |

Set Speaker Volume

void setSpeakerVolume(int volume);

| Name | Description |
|--------|--|
| volume | Set speaker volume from 0 (min) to max (255) |

Set Microphone Volume

void setMicrophoneVolume(int volume);

| Name | Description |
|--------|---|
| volume | Set microphone volume from 0 (min) to max (255) |

IAgoraAudioEventHandler Interface Methods

The following callback methods, within `IaudioEventHandler`, are executed upon events such as: joining a call, error reports, call quality reports, etc.

onLoadAudioEngineSuccess

`virtual void onLoadAudioEngineSuccess() = 0;`

You should implement this method to indicate what happens when the Agora audio engine is loaded correctly. This means the app was able to connect to an available audio server. From this point the audio engine is working, meaning it

is in communication mode. This is the ideal location in your code to create a time marker, allowing you to calculate and display call duration.

onJoinSuccess

```
virtual void onJoinSuccess(const char* channel, uid_t uid, int elapsed) = 0;
```

Indicates that the server has authenticated the client. The channel id is based on the channel name parameter specified by the join() API method. If the user id was not specified with the call to join(), the server will automatically allocate one.

| Name | Description |
|---------|-------------|
| channel | Channel id |
| uid | User id |
| elapsed | Delay in ms |

onRejoinSuccess

```
virtual void onRejoinSuccess(const char* channel, uid_t uid, int elapsed) = 0;
```

Indicates that the client has rejoined the server. Like onJoinSuccess(), the channel id and user id are passed back.

| Name | Description |
|---------|-------------|
| channel | Channel id |
| uid | User id |
| elapsed | Delay in ms |

onError

```
virtual void onError(int rescode, const char* msg) = 0;
```

Indicates that an error occurred in the audio engine or from the network connection.

| Name | Description |
|---------|---|
| rescode | <p>EVENT_LOAD_AUDIO_ENGINE_ERROR = 1001: failed to initialize audio engine</p> <p>EVENT_START_CALL_ERROR = 1003: failed to start audio engine. Typically this is caused because the audio device is in use by another app</p> <p>EVENT_JOIN_GET_AUDIO_ADDR_TIMEOUT = 11002:</p> |

| | |
|-----|--|
| | <p>voice server list timeout</p> <p>EVENT_JOIN_GET_AUDIO_ADDR_FAILED = 11003: error code received when requesting voice server list</p> <p>EVENT_JOIN_GET_AUDIO_ADDR_ZERO_ADDR = 11004: The Voice Center Server (acts as a gateway, like DNS) response that a voice server is not available</p> <p>EVENT_JOIN_CONNECT_MEDIA_TIMEOUT = 12002: connection to voice server timeout</p> <p>EVENT_JOIN_LOGIN_MEDIA_TIMEOUT_ALL = 13003: login to voice server timeout</p> <p>EVENT_JOIN_LOGIN_MEDIA_FAILED = 13004: Failed to login to voice sever, server ACKed with error code</p> <p>EVENT_JOIN_LOGIN_REGET_AUDIO_ADDR = 13005: the voice central server (acts as a gateway, like DNS) tried all available voice servers but none are able to accept this call</p> |
| msg | Error message that you want to display to the end-user or use for other purposes |

onAudioQuality

virtual void onAudioQuality(unsigned int uid, int quality, unsigned short delay, unsigned short jitter, unsigned short lost, unsigned short lost2) = 0;

Reports audio quality. This callback function will be triggered every 2seconds.

| Name | Description |
|---------|---|
| uid | User id, i.e. the caller. |
| quality | <p>Voice quality rating</p> <p>MEDIA_QUALITY_EXCELLENT = 1</p> <p>MEDIA_QUALITY_GOOD = 2</p> <p>MEDIA_QUALITY_POOR = 3</p> <p>MEDIA_QUALITY_BAD = 4</p> |

| | |
|--------|---|
| | MEDIA_QUALITY_VBAD = 5 MEDIA_QUALITY_DOWN = 6 |
| delay | Voice delay in ms |
| jitter | Jitter is the delay of inbound packets, due to network congestion or queuing issues |
| lost | Packet loss ratio |
| lost2 | Number of times that 2 consecutive packets were lost |

onNetworkQuality

```
virtual void onNetworkQuality(int quality) = 0;
```

Reports network quality. This callback function will be triggered every 2 seconds.

| Name | Description |
|---------|--|
| quality | Network quality rating MEDIA_QUALITY_EXCELLENT = 1 MEDIA_QUALITY_GOOD = 2 MEDIA_QUALITY_POOR = 3 MEDIA_QUALITY_BAD = 4 MEDIA_QUALITY_VBAD = 5 MEDIA_QUALITY_DOWN = 6 |

onLeaveChannel

```
virtual void onLeaveChannel(const SessionStat& stat) = 0;
```

Indicates the user left the channel. This callback provides session statistics information including call duration and tx/rx bytes.

| Name | Description |
|------|--|
| stat | struct SessionStat { unsigned int duration; unsigned int txBytes; unsigned int rxBytes; }; |

onUserJoined

```
virtual void onUserJoined(uid_t uid, int elapsed) = 0;
```

Indicates that a particular user joined. This callback provides the user's id and join delay in ms.

| Name | Description |
|---------|-------------|
| uid | User id |
| elapsed | Delay in ms |

onUserOffline

```
virtual void onUserOffline(uid_t uid) = 0;
```

Indicates that a particular user has left the call.

| Name | Description |
|------|-------------|
| uid | User id |

onUpdateSessionStats

```
virtual void onUpdateSessionStats(const SessionStat& stat) = 0;
```

This callback provides updated session statistics information. This callback function will be triggered every 2 seconds.

| Name | Description |
|------|--|
| stat | <pre>struct SessionStat { unsigned int duration; unsigned int txBytes; // transmission unsigned int rxBytes; // receipt };</pre> |

onAudioEngineEvent

```
virtual void onAudioEngineEvent(int evt) = 0;
```

This callback provides audio engine event messages.

| Name | Description |
|------|---|
| evt | <pre>enum AUDIO_ENGINE_EVENT_CODE { AUDIO_ENGINE_RECORDING_ERROR = 0, // recording cannot proceed</pre> |

| | |
|--|--|
| | <pre> AUDIO_ENGINE_PLAYOUT_ERROR = 1, // player cannot proceed AUDIO_ENGINE_RECORDING_WARNI NG = 2, // other recorder related events AUDIO_ENGINE_PLAYOUT_WARNING = 3 // other player related events }; </pre> |
|--|--|

onAudioDeviceStateChanged

```
virtual void onAudioDeviceStateChanged(const char* deviceId, int deviceType,
int deviceState) = 0;
```

Indicates that the system's audio device state has changed, such as earphones unplugged from the device

| Name | Description |
|-------------|--|
| deviceId | device id identifying an audio device |
| deviceType | <pre> enum AUDIO_DEVICE_TYPE { UNKNOWN_AUDIO_DEVICE = -1, PLAYOUT_DEVICE = 0, RECORDING_DEVICE = 1 }; </pre> |
| deviceState | <pre> enum AUDIO_DEVICE_STATE_TYPE { AUDIO_DEVICE_STATE_ACTIVE = 1, AUDIO_DEVICE_STATE_DISABLED = 2, AUDIO_DEVICE_STATE_NOT_PRESENT = 4, AUDIO_DEVICE_STATE_UNPLUGGED = 8 }; </pre> |