

# **GENESYS**

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# SIP Endpoint SDK Developer's Guide

SIP Endpoint SDK Video Support

# SIP Endpoint SDK Video Support

This article discusses the video support that has been added to SIP Endpoint SDK. This support consists of:

- A new video control interface
- · New video call control features that have been added to the existing call control interface

This article also outlines the requirements for processing High Definition video.

## High Definition Video Requirements

## **Important**

If you plan on supporting High Definition (HD) video, please take the following requirements into account.

Due to the complex processing required for High Definition video, all of the endpoints that are involved in a video conversation must run on computers that meet a certain minimum hardware performance level. As the actual CPU performance can no longer be accurately measured in MHz and since video processing performance depends on a wide variety of factors, Genesys recommends that you use the free benchmarking tool NovaBench to assess whether your hardware meets the requirements for successful HD video processing.

For 720p HD video, the **minimum** requirements are (in addition to what the camera manufacturer requires):

- A total NovaBench score of at least 500 (with a Graphic Test sub-score of at least 12), with recommended scores of at least 800 and 20, respectively
- 2 GB RAM
- 1 Mbps upload and download speed (for a total bandwidth of 2 Mpbs)

SIP Endpoint SDK for .NET does not currently support video resolutions higher than 720p.

## Video Control

The IVideoControl interface is in the Genesyslab.Sip.Endpoint.Provider.CP namespace. It allows you to work with events and information related to:

Video source

#### Frame events

These features are discussed in the following sections.

#### Video Source

This SDK provides methods to get or set video source-related information:

Method	Description
bool HasCaptureDevice();	Returns true if the SDK has been configured to work with a camera.
String ^ GetCurrentCaptureDeviceName();	Returns the names of the devices currently being used for video.
array <videodevice^>^ GetAllSystemCaptureDevices();</videodevice^>	Returns the names of all available cameras known to the operating system.
GsStatus SetCaptureDevice(int deviceId);	Picks up a capture device by its device ID and gets its status if the device has been successfully reserved.
VideoCapability^ GetVideoCapability(int deviceId, int N);	Returns capture device video capability by device ID and capability order number.
int GetNumberVideoCapabilities(int deviceId);	Returns a number of supported capabilities by device ID.

#### Video Frame Events

The SDK also provides video frame-related events:

Method	Description
<pre>event EventHandler<endpointeventargs^>^ VideoFrameSizeReceived;</endpointeventargs^></pre>	Provides notification about video frame size.
<pre>event EventHandler<endpointeventargs^>^ VideoFrameDelivered;</endpointeventargs^></pre>	Provides notification that a video frame has been received with size shown by VideoFrameSizeReceived.

## Video Call Control

SIP Endpoint SDK's ICallControl interface, which is in the Genesyslab.Sip.Endpoint.Provider.Genesys namespace, has methods to dial and answer a session with or without video:

Method	Description
<pre>void Dial(int connectionId, String^ destination, bool video, String^ data);</pre>	Starts dialing to place an outgoing call from the connection with ID = connectionId to destination. If video=true, the video is offered.
void Answer(int sessionId, bool video);	Answers an incoming call with sessionId and if

Method	Description
	video=true, the offered video is accepted.

### Access to Raw Video Frames

SIP Endpoint SDK 8.1.2 for .NET provides access to raw video frames in BGR32 format.

The following API has been added:

```
public enum class VideoRenderFormat {
        i420
                 = 0,
                 = 1,
        YV12
        YUY2
                 = 2,
        UYVY
                 = 3,
        IYUV
                 = 4,
        ARGB
        RGB24
                 = 6,
                = 7,
        RGB565
        ARGB4444 = 8.
        ARGB1555 = 9,
       MJPEG = 10.
                = 11,
        NV12
        NV21
                = 12,
                 = 13,
        BGRA
        Unknown = 99
};
```

You can use these methods to add and remove local and remote videos:

The following methods may be used to add or remove the local or remote video renderer:

```
// Add/Remove External Local Video Renderer
void AddLocalVideoRenderer(VideoRenderFormat format);
void RemoveLocalVideoRenderer();

// Add/Remove External Remote Video Renderer
void AddRemoteVideoRenderer(VideoRenderFormat videoFormat, int sessionId);
void RemoveRemoteVideoRenderer(int sessionId);
```

## Adding Video to an Audio-Only Call

In order to add video to an existing session, one of the supported video codecs should be specified in the **SipEndpoint.config** file:

```
<domain name="codecs">
```

If **auto\_accept\_video** is enabled and the other party adds video during an active audio call, video is automatically added to the endpoint. However, if **auto\_accept\_video** is not enabled and the other party adds video during an active audio call, then the user is prompted to either accept or reject the video.

#### API

The ICallControl Interface has been extended with these methods:

```
GsStatus SendVideoOffer(int sessionId);
GsStatus AnswerVideoOffer(int sessionId, bool accept);
GsStatus RemoveVideoStream(int sessionId);
```

You can add video to audio-only calls using both built-in and external video frame renderers.

#### **Application Notes**

Local video should be started before you add the video. If the media offer is accepted at this point, the video becomes bidirectional. If there is a request to remove the video, the local video is disconnected from the session.

Audio-video scenarios

#### **Video Offer Rejected By Remote Party**

Local party	Remote party	API
Establish audio-only session		
Voice to video escalation		
Start local video		
Request to add video		SendVideoOffer(sessionId)
	Receive session status	SessionStatus.MediaOffer
	Reject video offer	<pre>AnswerVideoOffer(sessionId, FALSE)</pre>
Receive session status		SessionStatus.MediaRejected
Audio-only session established		

#### **Video Offer Accepted By Remote Party**

Local party	Remote party	API
Add video to audio-only call		
Voice to video escalation		
Start local video		StartLocalVideo()
Request to add video		SendVideoOffer(sessionId)
	Receive session status	SessionStatus.MediaOffer
	Accept video offer	AnswerVideoOffer(sessionId, TRUE)
Receive session status		SessionStatus.MediaAccepted
Show remote video	Show remote video	StartRemoteVideo(sesionId)
Bidirectional video session established		

## **Video Removed From The Call**

Local party	Remote party	API
Existing bidirectional video session		
Remove video		RemoveVideoStream(sessionId)
Receive session status	Receive session status	SessionStatus.MediaRejected
Close remote video	Close remote video	StopRemoteVideo(sesionId)
Voice only session		