

# **GENESYS**

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## SIP Endpoint SDK Deployment Guide

New In This Release

### New In This Release

Check out the new features that have been added in the latest releases of SIP Endpoint SDK for Apple OS.

#### New in Release 8.1.2 for Apple OS

- SIP Endpoint SDK 8.1.2 for Apple OS supports OS X Mountain Lion (10.8)
- This release of SIP Endpoint SDK for Apple OS supports the following codecs:
  - G.711 (PCMA, PCMU)
  - G.722
  - iLBC (Internet Low Bitrate Codec)
  - iSAC (Internet Speech Audio Codec)
  - VP8 video
- SIP Endpoint SDK 8.1.2 for Apple OS supports the following Plantronics headset models in the standard headset mode:
  - Savi W440
  - Savi W7xx
  - Blackwire C320

SIP Endpoint SDK 8.1.2 for Apple OS also supports the following features:

- TLS 1.2 protocol (RFC 6176)
- · AGC (Automatic Gain Control)
- MWI (Message Waiting Indicator)
- · You can now specify these behaviors when a SIP Endpoint user does not have a working USB headset:
  - Whether SIP Endpoint should automatically reject an incoming call
  - The SIP error code to be sent to the inviting party
- INVITE messages now have an additional header that contains user data. This data can be obtained by using the following new method of GSSessionControlService:

```
- (GSResult) dialFrom:(id<GSConnection>)connection
to:(NSString*)destination withData:(NSString *)data;
```

- · Hangup on RTP inactivity timeout
- · Configuration of:
  - RTP port ranges

- SIP port ranges
- · Continuous Registration
- · DTMF tones can now be sent using SIP INFO
- VoIP monitoring of packet jitter and latency using RTP statistics
- NAT (Network Address Translation) traversal methods:
  - ICE (Interactive Connectivity Establishment)
  - STUN (Session Traversal Utilities for NAT)
  - TURN (Traversal Using Relay NAT)
- Connection to Genesys SIP Cluster

In addition, this release of SIP Endpoint SDK for Apple OS uses a SIP stack that has been developed by Genesys.

#### New in Release 8.1.1 for Apple OS

- SIP Endpoint SDK 8.1.1 for Apple OS supports OS X Lion (10.7)
- This release of SIP Endpoint SDK for Apple OS supports the following voice codecs:
  - G.711 (PCMA, PCMU)
  - G.722
  - iLBC (Internet Low Bitrate Codec)
- When devices and services use multiple voice codecs, the SIP Endpoint SDK for Apple OS supports the negotiation of the voice codec that will be used between them.
- The SIP Endpoint SDK for Apple OS supports Quality of Service (QoS), which helps guarantee that packet traffic for a voice or other media connection will not be delayed or dropped due to interference.
- The SIP Endpoint SDK for Apple OS supports additional security signaling and media encryption via SRTP (Secure Real-Time Transport Protocol).
- The SIP Endpoint SDK for Apple OS supports both first party call control (1PCC) and third party call control (3PCC).