GENESYS™

Integration Reference Manual

SIP Server 8.1 Integration Reference

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Welcome to the *SIP Server 8.1 Integration Reference*. This document introduces you to the concepts, terminology, and procedures related to integrating SIP Server with SIP endpoints, softswitches, and gateways. The reference information includes, but is not limited to, configuration options, limitations, and switch-specific functionality. This document is designed to be used along with the *SIP Server Deployment Guide*.

SIP Endpoint Application Notes

Polycom SIP Phones

CounterPath Bria SIP Phones

Yealink SIP Phones

AudioCodes/Genesys SIP Phones

Grandstream GXP SIP Phones

Grandstream GXP1625/GCC1700 SIP

SBC Application Notes

Sonus 1000/2000 SBC

Sonus 5000 SBC

AudioCodes Mediant SBC

Oracle Enterprise SBC

Oracle Acme Packet E-SBC Version 8.3.x

CUBE SBC

Switch Integrations

Siemens OpenScape Voice

Cisco Unified Communications Manager

Alcatel OXE

BroadSoft BroadWorks

RedSky E911 Manager

SIP Trunks Application Notes

AT&T IP Toll Free with AudioCodes Mediant SBC

AT&T IP Flexible Reach with AudioCodes Mediant SBC

AT&T IPTF and IPXC with Sonus SBC GSX/PSX

Colt SIP Trunk with AudioCodes Mediant SBC

Telenor SIP Trunk with AudioCodes Mediant SBC

Windstream SIP Trunk with AudioCodes Mediant SBC

CenturyLink SIP Trunk with AudioCodes

Media Gateway Application Notes

AudioCodes Mediant Gateway

Cisco Media Gateway

Network Load Balancer Integrations

F5 Networks BIG-IP LTM

SIP Endpoint SDK

SIP Endpoint SDK