

GENESYS

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Integration Reference Manual

Asterisk

Asterisk

This topic describes how to integrate SIP Server with the Asterisk switch. It contains the following sections:

- Overview
- · Asterisk for Business Calls Routing
- Asterisk as a Voicemail Server
- · Asterisk as a Media Server

Note: The instructions in this topic assume that both Asterisk and SIP Server are fully functional as stand-alone products. The instructions only highlight modifications to the existing configuration to make these products work as an integrated solution.