Integration Reference Manual

Siemens OpenScape Voice

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Siemens OpenScape Voice

This topic describes how to integrate SIP Server with the Siemens OpenScape Voice. It contains the following sections:

- Overview
- Configuring OpenScape Voice
- Configuring DN Objects
- Support for First-Party Call-Control Operations
- Support for Split-Node Deployments
- Handling Call Forwarding Loop

Note: The instructions in this topic assume that OpenScape Voice is fully functional and is routing calls before Genesys products are installed. They also assume that SIP Server has already been configured to function properly in stand-alone mode, and that configuration between SIP Server and Universal Routing Server (URS) has already been completed.