

# **GENESYS**

This PDF is generated from authoritative online content, and is provided for convenience only. This PDF cannot be used for legal purposes. For authoritative understanding of what is and is not supported, always use the online content. To copy code samples, always use the online content.

## SIP Server Deployment Guide

Remote Agents Support

## Contents

- 1 Remote Agents Support
  - 1.1 Configuring remote agents located behind the softswitch
  - 1.2 Configuring remote agents located behind the SBC/gateway
  - 1.3 Remote agents with non-provisioned phone numbers

# Remote Agents Support

SIP Server supports remote agents that use legacy PSTN phones. These agents could be working from their homes, or in a branch office that has simple PSTN connectivity.

SIP Server supports the following configurations for remote agents, depending on the remote agent locations:

- Remote agents located behind the softswitch
- Remote agents located behind the SBC/gateway
- Remote agents with non-provisioned phone numbers

To learn about benefits of nailed-up connections and how to configure them, refer to the Nailed-Up Connections for Agents topic.

To reconfigure office-based agents to their remote home-based locations, refer to the Enabling officebased agents to work from home topic.

For general information about configuring SIP devices, refer to the Configuring Devices and Services section in the *SIP Server Deployment Guide*.

### Configuring remote agents located behind the softswitch

Remote agent location	VOIP Service DN: Softswitch configuration	Extension DNs configuration
Behind the softswitch	<pre>[TServer]     contact = <the communication="" contact="" for="" server="" sip="" softswitch="" that="" the="" uri="" uses="" with="">     prefix = <the a="" be="" characters="" for="" initial="" match="" must="" number="" of="" particular="" selected="" softswitch="" that="" the="" to="">     service-type =     softswitch     refer-enabled = false     dual-dialog-enabled =     false</the></the></pre>	The Extension DN (Number property) for each remote agent must be configured with the PSTN number—for example, +1 555 123 1111, and contain no options. [TServer] <no options=""></no>

Remote agent location	VOIP Service DN: Softswitch configuration	Extension DNs configuration
	<ul> <li>reject-call-notready = true (recommended, not mandatory)</li> <li>sip-cti-control = <ensure option<br="" that="" this="">is not configured&gt;</ensure></li> </ul>	
With nailed-up connections behind the softswitch	<pre>[TServer]     contact = <the communication="" contact="" for="" server="" sip="" softswitch="" that="" the="" uri="" uses="" with="">     prefix = <the a="" be="" characters="" for="" initial="" match="" must="" number="" of="" particular="" selected="" softswitch="" that="" the="" to="">     service-type =     softswitch     refer-enabled = false     dual-dialog-enabled =     false     reject-call-notready =     true (recommended, not     mandatory)     sip-cti-control =     <ensure configured="" is="" not="" option="" that="" this=""> </ensure></the></the></pre>	The Extension DN (Number property) for each remote agent must be configured with the PSTN number—for example, +1 555 123 1111, and contain the following options: [TServer] • line-type = 1

## Configuring remote agents located behind the SBC/gateway

Remote agent location	Extension DNs configuration
Behind the SBC/gateway	The Extension DN (Number property) for each remote agent must be configured with the PSTN number—for example, +1 555 123 1111, and contain the following options: [TServer]

Remote agent location	Extension DNs configuration
	<ul> <li>contact = <the <br="" contact="" of="" pstn="" sbc="" the="" uri="">gateway, depending on the agent location&gt;</the></li> </ul>
	<ul> <li>refer-enabled = false</li> </ul>
	<ul> <li>dual-dialog-enabled = false</li> </ul>
	<ul> <li>reject-call-notready = true (recommended, not mandatory)</li> </ul>
	<ul> <li>sip-cti-control = <ensure is<br="" option="" that="" this="">not configured&gt;</ensure></li> </ul>
	The Extension DN (Number property) for each remote agent must be configured with the PSTN number—for example, +1 555 123 1111, and contain the following options:
	[TServer]
	<ul> <li>contact = <the contact="" of="" pstn<br="" the="" uri="">gateway/SBC, depending on the agent location&gt;</the></li> </ul>
With nailed-up connections behind the SBC/ gateway	• refer-enabled = false
	<ul> <li>dual-dialog-enabled = false</li> </ul>
	<ul> <li>reject-call-notready = true (recommended, not mandatory)</li> </ul>
	<ul> <li>sip-cti-control = <ensure is<br="" option="" that="" this="">not configured&gt;</ensure></li> </ul>
	• line-type = 1

#### Limitations

Due to the specifics of gateway behavior in performing SIP REFER methods, support for remote agents has some limitations. In order to use remote agents, you must perform one of the two following steps:

- Provision customers and remote agents to use physically separate gateways (otherwise, calls from
  agents to customers take shortcuts within gateways, which means that SIP Server loses track of the call
  and therefore cannot perform call control). Even in this configuration, direct calls between two remote
  agents on the same gateway are not visible to SIP Server.
- Disable the SIP REFER method for the gateways where the remote agents are located. This enables SIP Server to see agent-to-customer and agent-to-agent calls.

### Remote agents with non-provisioned phone numbers

Starting with version 8.1.102.93, SIP Server improves provisioning of remote agent DNs in the Configuration Database. It is no longer required to provision external phone numbers (for example, agent's PSTN numbers) in the Configuration Database. You must create an Extension DN for each remote agent where a DN number can be a primary office DN number or any other number if an agent doesn't have a primary office DN.

The external phone number is used to reach the agent during the agent session only, thereby limiting the lifetime of the external phone number to a particular agent session. In other words, after the agent is logged out, any associations with that external phone number are removed.

The non-provisioned phone number to be used for the agent session is passed to SIP Server in the TAgentLogin request in AttributeExtensions as the **agent-phone** key. AttributeThisDN of that request will contain the agent DN configured in the Configuration Database.

This feature requires Workspace Web Edition (WWE) version 8.5.201.95 or later.

#### Configuring remote agents with non-provisioned phone numbers

Remote agent location	VOIP Service DN: Softswitch configuration	Extension DNs configuration
Behind the softswitch	<pre>[TServer]     contact = <the communication="" contact="" for="" server="" sip="" softswitch="" that="" the="" uri="" uses="" with="">     prefix = <the a="" be="" characters="" for="" initial="" match="" must="" number="" of="" particular="" selected="" softswitch="" that="" the="" to="">     service-type =     softswitch     refer-enabled = false     dual-dialog-enabled =     false     reject-call-notready =     true (recommended, not     mandatory)     sip-cti-control =     <ensure configured="" is="" not="" option="" that="" this=""> </ensure></the></the></pre>	<b>[TServer]</b> <no options=""></no>
With nailed-up connections behind the	[TServer]	[TServer]

Remote agent location	VOIP Service DN: Softswitch configuration	Extension DNs configuration
softswitch	<ul> <li>contact = <the contact<br="">URI that SIP Server uses for communication with the softswitch&gt;</the></li> <li>prefix = <the initial<br="">characters of the number that must match a particular softswitch for that softswitch to be selected&gt;</the></li> <li>service-type = softswitch</li> <li>refer-enabled = false</li> <li>dual-dialog-enabled = false</li> <li>reject-call-notready = true (recommended, not mandatory)</li> <li>sip-cti-control = <ensure option<br="" that="" this="">is not configured&gt;</ensure></li> </ul>	<ul> <li>line-type = 1</li> <li>connect-nailedup-on-login = gcti::park</li> </ul>

#### Limitations

- If a non-provisioned phone number is used for the agent session, the agent can only initiate calls using the agent desktop. 1pcc calls originated from the non-provisioned phone number are not supported.
- For agents with nailed-up connections that use a non-provisioned number for the agent session, an establishment of the nailed-up connection by calling into a contact center routing point is not supported.
- Hunt Groups in Business Continuity (BC) functionality are not supported by this feature. That is, in the BC deployment, agent logging with a non-provisioned external phone number to a DN that is a member of the Hunt Group is not supported.