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# Integration Reference Manual

Configuring OpenScape Voice

12/13/2025

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# Configuring OpenScape Voice

This page provides an overview of the main steps that are required to configure OpenScape Voice. Complete all steps in the order in which they are listed.

1. Check that OpenScape Voice is working
2. Configure Numbering Plans
3. Configure the Endpoint Profile
4. Configure the Endpoint
5. Configure Gateway Destinations
6. Configure Prefix Access Codes
7. Configure Destination Codes
8. Configure Agent Destinations
9. Configure Agent Access and Destination Codes
10. (Optional) Configure Click-to-Answer
11. (Optional) Configure emergency call routing

## 1. Check Minimum Functionality in OpenScape Voice

The procedures in this topic assume that OpenScape Voice is functional and routing calls appropriately. There should already be at least one Numbering Plan that has gateways and nonagent subscribers in it. For more information, see Siemens OpenScape Voice-specific documentation.

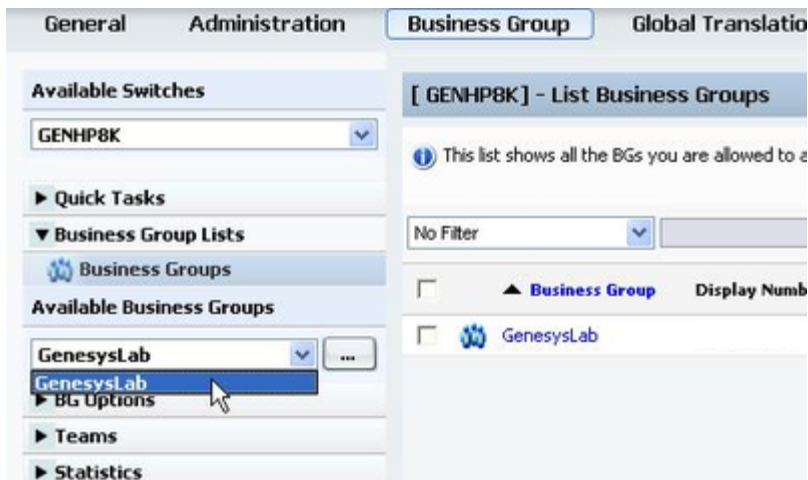
## 2. Configuring Numbering Plans

The instructions in this topic assume that OpenScape Voice is functional and routing calls appropriately. There should already be at least one Numbering Plan with configured gateways and nonagent subscribers.

**Purpose:** To create the Numbering Plans that will contain the Agents and SIP Server.

### Start

1. Log in to the HiPath Assistant, and navigate to the Business Group of the contact center that you want to configure--for example, GenesysLab.



Selecting the Business Group

2. Click Private Numbering Plans.



Selecting Private Numbering Plans

3. In the Private Numbering Plans dialog box, click Add.
4. Add two new Private Numbering Plans: one for your agents and one for SIP Server itself--for example, Agents and SIPServer, respectively.



Creating Private Numbering Plans

When you are finished, the dialog box shown in the following figure appears.

	Agents	0	User-defined	Private
	Gen	0	User-defined	Private
	SIPServer	0	User-defined	Private

Private Numbering Plans

**End**

### 3. Configuring a SIP Server Endpoint Profile

**Start**

1. Click Private Numbering Plan, and then click the SIP Server Numbering Plan—for example, SIPServer.



Selecting the Numbering Plan

2. Click Endpoint Management, and then click Endpoint Profiles.



Selecting Endpoint Profiles

3. In the Endpoint Profile: <Business Group> dialog box on the General tab, enter a name for this configured Endpoint Profile in the Name text box. This will associate the endpoint that uses it with the Numbering Plan in which the Endpoint Profile was created.



[ GENHPBK ] - Endpoint Profile: GenesysLab - ""

General Endpoints Services Blocked Numbers

Enter the profile data.

**Endpoint Profile**

Please enter a unique name to identify this profile.

Name: SIPServerEPP

Remark:

Business Group: GenesysLab

Numbering Plan: SIPServer

**Management Information**

Please enter the data for the following fields in the corresponding screens.

Class of Service:

Routing Area:

Calling Location:

SIP Privacy Support: Basic

Failed Calls Intercept Treatment: Disabled

Language: English

Save Cancel

Configuring an Endpoint Profile

- (Optional) If there are existing dialing rules and conventions that require the use of Class of Service and Routing Areas, enter that information. As a general rule, give this Endpoint Profile the same calling access as you would give to your agents
- When you are finished, click Save.
- In the Endpoint Profile: <Business Group> dialog box on the Services tab, enable the Call Transfer service, by selecting Yes from the drop-down menu.



General Endpoints Services

Voice mail: No

Call Transfer: Yes

Call Forward Invalid Destination: No

Enabling the Call Transfer Service

**End**

## 4. Configuring a SIP Server Endpoint

### Start

- Click Private Numbering Plan, and then click the SIP Server Numbering Plan—for example, SIPServer.
- Click Endpoints, and then click Add.



Selecting Endpoints

3. In the Endpoint: <Business Group> dialog box, click the General tab, and do the following:
  - a. In the Name text box, enter a unique name for this configured Endpoint.
  - b. Select the Registered check box.
  - c. Set the Profile text box to the Endpoint Profile that you created for SIP Server, by clicking the browse (...) button.



Configuring Endpoints: General Tab

4. In the Endpoint: <Business Group> dialog box, click the SIP tab, and do the following:
  - a. Make sure that the Type text box is set to Static.
  - b. In the Endpoint Address text box, enter the IP address of SIP Server.
  - c. From the Transport protocol drop-down box, select UDP or TCP, depending on SIP Server.

[GENOSY] - [GenosysLab] - [Main Office] - Add Endpoint

General SIP **Attributes** Aliases Routes Accounting

Endpoint Type

SIP Private Networking: ☐

SIP Trunking: ☒

SIP-Q Signaling: ☐

SIP Signaling

For the static Endpoints the address of the SIP signaling interface can be specified in IP or FQDN format.  
Note that the address of the signaling interface cannot be modified unless the entry in the security section has first been removed.

Type:

Signaling Address Type:

Endpoint Address:

Port:

Transport protocol:

Configuring Endpoints: SIP Tab

5. Click the Attributes tab, and do the following:
  - a. Select the Transfer HandOff check box.  
There is a known limitation of the Transfer HandOff feature. The full number must be used to transfer a call when this feature is activated.
  - b. Select the Do not Send Invite without SDP check box.
  - c. When you are done, click Save.

Allow Sending of Insecure Referred-By Header ☐

Override SRM Codec Restriction ☐

Transfer HandOff ☒

Send P-Preferred-Identity rather than P-Asserted-Identity ☐

Send domain name in From and P-Preferred-Identity headers ☐

Send Redirect Number instead of calling number for redirected calls ☐

Do not send Diversion header ☐

Do not Send Invite without SDP ☒

Send URI in Telephone Subscriber Format ☐

Rerouting Direct Incoming Calls ☐

Save Cancel

Configuring Endpoints: Attributes Tab

6. Click the Aliases tab, and then click Add.
7. In the Alias dialog box, do the following:
  - a. In the Name text box, enter the IP address that you entered in the Endpoint Address text box in Step 4.
  - b. Unless you have OpenScape Voice version 5 and later, set the Type text box to SIP URL. (This is done automatically in version 5.)
  - c. Click OK.





Configuring Endpoints: Aliases Tab

8. In the Endpoint dialog box, click Save.
9. When the confirmation message box appears, informing you that the Endpoint was created successfully, click Close.

### End

## 5. Configuring SIP Server Destinations for Gateways

**Purpose:** To create Gateway Destinations for SIP Server to route calls. The Endpoints of such Gateway Destinations must already be configured in OpenScape Voice. SIP Server routes calls to Gateways and to phones. Because calls to the phones are routed via the E.164 Numbering Plan, no Destinations have to be configured for them.

### Start

1. Click Private Numbering Plan, and then click the SIP Server Numbering Plan—for example, SIPServer.
2. Click Destinations and Routes, then Destinations, and then click Add.



Selecting Destinations

3. In the Destination dialog box, on the General tab, do the following:
  - a. In the Name text box, enter a unique name for the Destination—for example, SIPServerGWDEST. The name must be unique within the switch configuration database.
  - b. Make sure that all check boxes are cleared.

- c. When you are finished, click Save.



Configuring a Gateway Destination

4. In the Destination - <Business Group> dialog box, click the Destination that you just created.
5. Click the Routes tab, and then click Add.
6. In the Route dialog box, do the following:
  - a. In the ID text box, enter 1 for this particular route.
  - b. Set the Type text box to SIP Endpoint.
  - c. Set the SIP Endpoint text box to the Endpoint that you created in [Configuring a SIP Server Endpoint](#) by clicking the browse (...) button, selecting the Numbering Plan that contains the Endpoint for the gateway to which you will be routing (for example, the general Numbering Plan), and then selecting the Endpoint.
  - d. Do not modify the digit string for calls that are being routed from SIP Server. All modifications to the digit string should be completed before the calls arrive to SIP Server.

**[ GENHP8K ] - Route**

A route connects the destination with an endpoint representing a gateway.

**ID**

The Route ID indicates the priority level.

ID:

Type:

SIP Endpoint:  ...

**Originator Attributes**

Restricts the traffic according to specified settings. Routes with the same restrictions can be prioritized.

Signaling Type:

Bearer Capability:

**Destination Directory Number**

Last chance to modify the dialed digits for the gateway.  
Number of digits to delete: Leading digits are cut off from the Directory Number.  
Digits to insert: the digit string is added to the beginning of the remaining digits.

Number of digits to delete:

Digits to insert:

Nature of Address:

Save Cancel

Configuring a Route for a Gateway Destination

- When you are finished, click Save.
- When the confirmation message box appears, informing you that the Route was added successfully, click Close.
- In the Destination dialog box, click OK. You will now be able to view the Route that you just created in the Routes dialog box.
- Repeat Steps 2-9 to create other gateway Destinations for SIP Server, as necessary.

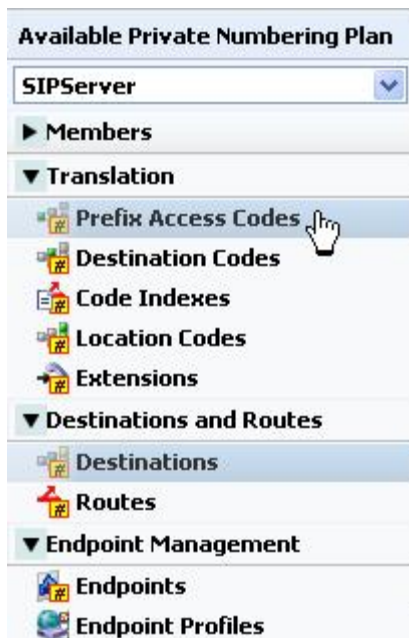
**End**

## 6. Configuring SIP Server Prefix Access Codes

**Purpose:** To configure Prefix Access Codes that SIP Server will dial to reach Subscribers and Gateways.

**Start**

- Click Private Numbering Plan, and then click the SIP Server Numbering Plan—for example, SIPServer.
- Click Translation, click Prefix Access Codes, and then click Add.



Selecting Prefix Access Codes

3. For calls that are to be routed to Subscribers: In the Prefix Access Code: <Business Group> dialog box, do the following:
  - a. In the Prefix Access Code text box, enter the digits you want to use to route calls to Subscribers.  
**Note:** For the SIP Server Numbering Plan, minimal modifications should be required. Dialed numbers should be modified before they reach SIP Server. This convention should be followed at all sites, to simplify the solution as much as possible.
  - b. Set the Prefix Type text box to Off-net Access.
  - c. Set the Nature of Address text box to Unknown.
  - d. Set the Destination Type text box to E164 Destination.
  - e. Click Save.

[ GENHPBK ] - Prefix Access Code : GenesysLab -

**General**    Destination Codes

**Identification and Modification**

If the dialed digits match this code, the specified modification to these dialed digits is executed.

Prefix Access Code: 12

Remark:

Minimum Length: 4

Maximum Length: 7

Digit Position: 0

Digits to insert:

**Settings**

Specify additional parameters to determine how the call will be routed.

Prefix Type: Off-net Access

Nature of Address: Unknown

Destination Type: E164 Destination

Service:

Save Cancel

Configuring a Prefix Access Code for Calls Routed to Subscribers

6. When the confirmation message box appears, informing you that the Prefix Access Code was created successfully, click Close.
7. If agents will be allowed to make external calls: In the Prefix Access Code dialog box, click Add again.
8. In the Prefix Access Code dialog box, do the following:
  - a. In the Prefix Access Code text box, enter the digits that you want to use to route calls to Gateways. The matched digits will be site-specific, and there should be minimal modification of the digit string.
  - b. Set the Prefix Type text box to Off-net Access.
  - c. Set the Nature of Address text box to Unknown.
  - d. Set the Destination Type text box to None, so you will be able to route the call from a Destination Code.
  - e. Click OK.

[ GENHPBK ] - Prefix Access Code : GenesysLab -

**General**    Destination Codes

**Identification and Modification**

If the dialed digits match this code, the specified modification to these dialed digits is executed.

Prefix Access Code: 34

Remark:

Minimum Length: 4

Maximum Length: 7

Digit Position: 0

Digits to insert:

**Settings**

Specify additional parameters to determine how the call will be routed.

Prefix Type : Off-net Access

Nature of Address: Unknown

Destination Type: None

Destination Name:

Save Cancel

Configuring a Prefix Access Code for Calls Routed to Gateways

6. When the confirmation message box appears, informing you that the Prefix Access Code was created successfully, click Close.

**End**

### Next Steps

Continue with the following procedure, unless calls are routed only to Subscribers:

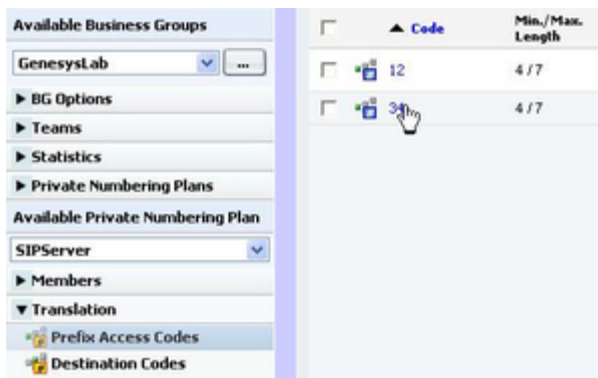
- [Configuring SIP Server Destination Codes](#)

## 7. Configuring SIP Server Destination Codes

**Purpose:** To configure SIP Server Destination Codes to route calls to non-Subscriber devices.

### Start

1. Click Private Numbering Plan, and then click the SIP Server Numbering Plan—for example, SIPServer.
2. Click Prefix Access Codes.
3. Click the Prefix Access Code that you created for non-Subscriber devices.



Selecting a Prefix Access Code

4. In the Prefix Access Code dialog box, click the Destination Codes tab.
5. In the Destination Code dialog box, do the following:
  - a. Set the Destination Type text box to Destination.
  - b. Set the Destination Name text box to the Destination that you created for SIP Server in [Configuring SIP Server Destinations for Gateways](#), by clicking the browse (...) button.

**[ GENHP8K ] - Destination Code - 34**

**General** | Extensions

**Identification**

This destination code will be used for a call if the dialed or modified (in PAC) digits and the Nature of Address are matching.

Destination Code: 34

Remark:

Country Code:

Nature Of Address: Unknown

Traffic Type: NONE

**Originator Attributes**

Optionally, an additional match is required if the originator of the call belongs to the specified Class of Service and Routing Area.

Class Of Service:

Routing Area:

NPA:

**Destination**

Specify additional parameters to determine how the call will be routed.

Destination Type: Destination

Destination Name: SIPServerGWDEST

DN Office Code:

Save Cancel

Configuring a Destination Code

6. Click Save.
7. When the confirmation message box appears, informing you that the Destination Code was created

successfully, click Close.

## End

## 8. Configuring an Agent Destination for SIP Server

**Purpose:** To configure a Destination for the Agent Numbering Plan for SIP Server.

### Start

1. Click Private Numbering Plan, and then click the Agent Numbering plan—for example, Agents.
2. Click Destinations and Routes, click Destinations, and then click Add.



Selecting Destinations

3. In the Destination - <Agent Numbering Plan> dialog box, click the General tab, and then do the following:

- a. In the Name text box, enter a unique name for the Destination.

**Note:** Destinations must be unique within the switch configuration database, not just within the Numbering Plan and Business Group.

- b. Make sure that all check boxes are cleared.
- c. When you are finished, click Save, and then close the dialog box.



Configuring a SIP Server Destination in the Agent Numbering Plan



4. Click the Destination that you just created—for example, SIPServer.

5. Click the Routes tab, and then click Add.

6. In the Route dialog box, do the following:

- a. In the ID text box, enter 1.

**Note:** The ID of the first Route must always be 1.

- b. Set the Type text box to SIP Endpoint.

- c. Set the SIP Endpoint text box to the Endpoint that you created for SIP Server in [Configuring a SIP Server Endpoint](#), by clicking the browse (...) button.

- d. When you are finished, click Save.

**Note:** Genesys recommends that you not modify the dialed-digit string that is passed on to SIP Server at this point.

[ GENHP8K ] - Route

A route connects the destination with an endpoint representing a gateway.

**ID**

The Route ID indicates the priority level.

ID:

Type:

SIP Endpoint:  ...

**Originator Attributes**

Restricts the traffic according to specified settings. Routes with the same restrictions can be prioritized.

Signaling Type:

Bearer Capability:

**Destination Directory Number**

Last chance to modify the dialed digits for the gateway.  
Number of digits to delete: Leading digits are cut off from the Directory Number.  
Digits to insert: the digit string is added to the beginning of the remaining digits.

Number of digits to delete:

Digits to insert:

Nature of Address:

Save Cancel

Configuring a Route for SIP Server in the Agent Numbering Plan

5. When the confirmation message box appears, informing you that the Route was added successfully, click Close.

**End**

## 9. Configuring Agent Prefix Access Codes and Destination Codes

In this section, you configure dialing patterns for the Agents. Every number that the agent dials must be configured. If an agent dials a four-digit extension, the Prefix Access Code should be configured to convert the dialed-digit string to the full E.164 code that OpenScape Voice expects. If the agent dials a number that must be routed to an external gateway, make sure that the dialed-digit string is correct for that gateway before it reaches SIP Server.

As mentioned earlier, all calls must go to SIP Server first; otherwise, the calls will not be visible to SIP Server. In the Private Numbering Plan for agents, every Prefix Access Code must route the call to a Destination Code that points the call to SIP Server. It is best to copy the nonagent Prefix Access Codes from the General Numbering Plan; however, make sure that the destination is always SIP Server.

### Start

1. Click **Private Numbering Plan**, and then click the Agent Numbering Plan—for example, **Agents**.
2. Click **Translation**, click **Prefix Access Codes**, and then click **Add**.
3. In the **Prefix Access Code** dialog box, do the following:
  - a. In the **Prefix Access Code** text box, enter the digits you want to use for routing, and any modifications that OpenScape Voice will need to make in order to route the call properly.
  - b. Set the **Prefix Type** text box to **Off-net Access**.
  - c. Set the **Nature of Address** text box to **Unknown**.
  - d. Set the **Destination Type** text box to **None**.
  - e. Click **Save**, and close the dialog box.

[ GENHP8K ] - Prefix Access Code : GenesysLab -

**General** Destination Codes

**Identification and Modification**

If the dialed digits match this code, the specified modification to these dialed digits is executed.

Prefix Access Code: 12

Remark:

Minimum Length: 4

Maximum Length: 7

Digit Position: 0

Digits to insert: 345

**Settings**

Specify additional parameters to determine how the call will be routed.

Prefix Type: Off-net Access

Nature of Address: Unknown

Destination Type: None

Destination Name:

Save Cancel

Configuring a Prefix Access Code for the Agent Numbering Plan

- f. In the Prefix Access Code dialog box, click the Prefix Access Code that you just created, and then click the Destination Codes tab.
7. In the Destination Code dialog box, click the General tab, and then do the following:
  - a. Do not modify the Destination Code text box.
  - b. Make sure that the Nature of Address text box is set to Unknown.
  - c. Make sure that the Destination Type text box is set to Destination.
  - d. Set the Destination Name text box to the Destination that you created for SIP Server in [Configuring an Agent Destination for SIP Server](#)—for example, SIPServer--by clicking the browse (...) button.
  - e. When you are finished, click Save.

**[ GENHP8K ] - Destination Code - 34512**

**General** | Extensions

**Identification**

This destination code will be used for a call if the dialed or modified (in PAC) digits and the Nature of Address are matching.

Destination Code: 34512

Remark:

Country Code:

Nature Of Address: Unknown

Traffic Type: NONE

**Originator Attributes**

Optionally, an additional match is required if the originator of the call belongs to the specified Class of Service and Routing Area.

Class Of Service:

Routing Area:

NPA:

**Destination**

Specify additional parameters to determine how the call will be routed.

Destination Type: Destination

Destination Name: SIPServer

DN Office Code:

Save Cancel

Configuring a Destination Code for the Agent Destination

- When the confirmation message box appears, informing you that the Destination Code was created successfully, click Close.
- Repeat Steps 2-6 to create other Prefix Access Codes and Destination Codes, as necessary.

**End**

## 10. (Optional) Configure Click-to-Answer

This configuration is not required for the integration to work, however, some might be required by local laws, or make the solution easier to configure.

**Purpose:** The Click-to-Answer feature enables agents to click within Genesys Agent Desktop to answer the phone. The Click-to-Answer feature requires the referenced Patchset on OpenScape Voice and a device that supports it. The current procedure provides instructions for OpenStage phones.

**Start**

1. On the phone that you have to configure, select Configuration.



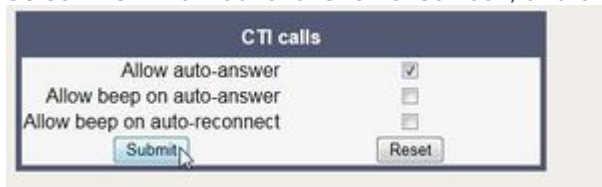
Selecting Configuration on the OpenStage Phone

2. Click Incoming calls, and then click CTI calls.



Configuring CTI Calls on the OpenStage Phone

3. Select the Allow auto-answer check box, and click Submit.



Submitting Allow auto-answer on the OpenStage Phone

4. Repeat Steps 1-3 for every agent phone on the switch.

**End**

## 11. (Optional) Configure emergency call routing

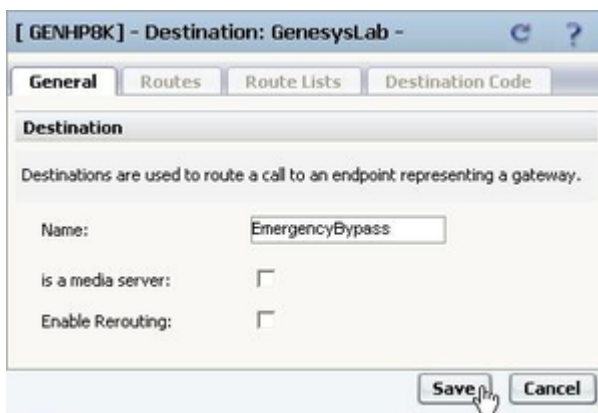
This configuration is not required for the integration to work, however, some might be required by local laws, or make the solution easier to configure.

The emergency call routing feature provides alternate call routing in cases in which SIP Server is unavailable, if your local emergency (or 911) laws require some form of alternate routing for agents.

During the first 30 seconds after the emergency calling support is activated, calls will fail to route. After that, OpenScape Voice will route calls via the alternate route that you configure and the calls will work.

### Start

1. Log in to the HiPath Assistant, and navigate to the Business Group of the contact center that you want to configure—for example, GenesysLab.
2. Click Private Numbering Plan, and then click the Agent Numbering Plan.
3. Click Destinations and Routes, click Destinations, and then click Add.
4. In the Destination dialog box, do the following:
  - a. In the Name text box, enter a new destination for the gateway through which you want emergency calls to go—for example, EmergencyBypass.
  - b. Make sure that all check boxes are cleared.
  - c. Click Save.



Configuring a Destination for Emergency Call Routing

4. Click the Destination that you just created—for example, EmergencyBypass.
5. Click the Routes tab, and then click Add. In this step you are adding a route that goes to SIP Server. This is necessary in order to prevent calls from bypassing SIP Server while it is working.
6. In the Route dialog box, do the following:
  - a. In the ID text box, enter 1. This route goes to SIP Server, just like all the others.
  - b. Set the Type text box to SIP Endpoint.
  - c. Set the SIP Endpoint text box to the Endpoint that you created in [Configuring a SIP Server Endpoint](#).
4. When you are finished, click Save.
5. Click the Destination that you just created—for example, EmergencyBypass.
6. Click the Routes tab, and then click Add again.
7. In the Route dialog box, do the following:
  - a. In the ID text box, enter 2.
  - b. Set the Type text box to SIP Endpoint.
  - c. Set the SIP Endpoint text box to the gateway for emergency calling.

- d. When you are finished, click Save.

**[ GENHPBK ] - Route**

A route connects the destination with an endpoint representing a gateway.

**ID**

The Route ID indicates the priority level.

ID:

Type:

SIP Endpoint:

**Originator Attributes**

Restricts the traffic according to specified settings. Routes with the same restrictions can be prioritized.

Signaling Type:

Bearer Capability:

**Destination Directory Number**

Last chance to modify the dialed digits for the gateway.  
Number of digits to delete: Leading digits are cut off from the Directory Number.  
Digits to insert: the digit string is added to the beginning of the remaining digits.

Number of digits to delete:

Digits to insert:

Nature of Address:

**Save** **Cancel**

Configuring a Route for Emergency Call Routing

5. Click Prefix Access Codes, and then click Add.
6. In the Prefix Access Code dialog box, do the following:
  - a. In the Prefix Access Code text box, enter the digits for your emergency number.
  - b. Set the Prefix Type text box to Off-net Access.
  - c. Set the Nature of Address text box to Unknown.
  - d. Set the Destination Type text box to None.
  - e. Click Save, and close the dialog box.

**[ GENHPBK ] - Prefix Access Code : GenesysLab -**

**General**    Destination Codes

**Identification and Modification**

If the dialed digits match this code, the specified modification to these dialed digits is executed.

**Prefix Access Code:**

**Remark:**

**Minimum Length:**

**Maximum Length:**

**Digit Position:**

**Digits to insert:**

**Settings**

Specify additional parameters to determine how the call will be routed.

**Prefix Type:**

**Nature of Address:**

**Destination Type:**

**Destination Name:**

Configuring a Prefix Access Code for Emergency Call Routing

6. In the Prefix Access Code dialog box, click the Destination Codes tab.
7. On the General tab, do the following:
  - a. Make sure that the Destination Type text box is set to Destination.
  - b. Set the Destination Name text box to the Destination that you created in Step 4—for example, EmergencyBypass—by clicking the browse (...) button.
  - c. When you are finished, click OK.



**[ GENHPBK ] - Destination Code - 911**

**General** | Extensions

**Identification**

This destination code will be used for a call if the dialed or modified (in PAC) digits and the Nature of Address are matching.

Destination Code: 911

Remark:

Country Code:

Nature Of Address: Unknown

Traffic Type: NONE

**Originator Attributes**

Optionally, an additional match is required if the originator of the call belongs to the specified Class of Service and Routing Area.

Class Of Service:

Routing Area:

NPA:

**Destination**

Specify additional parameters to determine how the call will be routed.

Destination Type: Destination

Destination Name: EmergencyBypass

DN Office Code:

Save Cancel

Configuring a Destination Code for Emergency Call Routing

**End**

### Next Steps:

- Configuration of OpenScape Voice is now complete. Proceed with [Configuring DN Objects](#).