



GUI Administration Guide for Cisco Unified SIP Proxy Release 10.2

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Americas Headquarters

Cisco Systems, Inc.
170 West Tasman Drive
San Jose, CA 95134-1706
USA
<http://www.cisco.com>
Tel: 408 526-4000
800 553-NETS (6387)
Fax: 408 527-0883

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CONTENTS

| | | |
|------------------|-----------------------|----------|
| CHAPTER 1 | Change History | 1 |
|------------------|-----------------------|----------|

| | | |
|------------------|--|----------|
| CHAPTER 2 | Welcome to Cisco Unified SIP Proxy | 3 |
| | Logging in to Cisco Unified SIP Proxy Graphical User Interface (GUI) | 3 |
| | About the Dashboard | 4 |
| | Commercial Open Source Licensing | 4 |

| | | |
|------------------|---|----------|
| CHAPTER 3 | Configuring SIP Stacks | 5 |
| | Viewing and Editing General Settings for SIP Stacks | 5 |
| | SIP Stack General Settings | 5 |
| | Adding and Deleting an Alias FQDN | 8 |
| | Adding and Deleting a Trusted Peer | 9 |

| | | |
|------------------|---|-----------|
| CHAPTER 4 | Configuring Networks | 11 |
| | Viewing a List of Networks | 11 |
| | Adding a Network | 11 |
| | Network Information | 12 |
| | Editing the General Settings for a Network | 13 |
| | Editing the SIP Retransmission Settings for a Network | 14 |
| | SIP Retransmissions | 15 |
| | Viewing and Deleting SIP Listen Points | 16 |
| | Adding a SIP Listen Point | 16 |
| | Editing the SIP Record-Route for a Network | 17 |

| | | |
|------------------|-------------------------------|-----------|
| CHAPTER 5 | Configuring Triggers | 19 |
| | Viewing and Deleting Triggers | 19 |

About Triggers 20

Example of a Trigger 20

Available Trigger Conditions and Cases 21

Adding a Trigger 26

Viewing, Adding, Moving, and Deleting Rules for a Trigger 27

Adding, Editing, and Deleting Conditions for a Trigger Rule 28

CHAPTER 6

Configuring Server Groups 31

Viewing a List of Server Groups 31

 Server Groups (Group Tab) Fields 32

Adding a Server Group 33

Editing a Server Group 34

Viewing and Editing the General Settings for All Server Groups 35

 Server Groups (General Settings Tab) Fields 35

Viewing and Deleting Server Group Elements 37

 Server Group (Elements Tab) Fields 38

Adding and Editing a Server Group Element 39

Viewing a List of SIP Ping Network Connections 39

 SIP Ping Fields 40

Adding a SIP Ping Configuration 41

Editing a SIP Ping Configuration 42

Viewing a List of Call Admission Control Endpoints 43

Changing the Limit of a Call Admission Control Endpoint 43

CHAPTER 7

Configuring Route Groups 45

Viewing a List of Route Groups and Corresponding Elements 45

 About Route Groups 46

 Route Group Fields 46

 Element Fields 47

Adding a Route Group 48

Viewing and Deleting Route Group Elements 49

Adding and Editing Route Group Elements 50

Editing a Route Group 50

| | | |
|------------------|--|-----------|
| CHAPTER 8 | Configuring Route Tables | 53 |
| | Viewing a List of Route Tables | 53 |
| | About Route Tables | 54 |
| | Route Table Fields | 54 |
| | Route Fields | 54 |
| | Adding a Route Table | 56 |
| | Viewing a List of Route Table Routes | 57 |
| | Adding a Route to a Route Table | 58 |
| | Exporting Active Routes | 59 |
| | Editing the Routes Associated with a Route Table | 59 |

| | | |
|------------------|---------------------------------------|-----------|
| CHAPTER 9 | Configuring Route Policies | 61 |
| | Viewing a List of Route Policies | 61 |
| | Route Policy Fields | 62 |
| | Route Policy Step Fields | 62 |
| | Adding a Route Policy | 64 |
| | Viewing a List of Route Policy Steps | 65 |
| | Adding or Editing a Route Policy Step | 66 |

| | | |
|-------------------|---|-----------|
| CHAPTER 10 | Configuring Normalization Policies | 69 |
| | Viewing a List of Normalization Policies | 69 |
| | About Normalization Policies | 70 |
| | Normalization Policy Fields | 70 |
| | Request URI, URI Component Fields | 70 |
| | Request URI, URI Conversion Fields | 71 |
| | Request URI, URI Parameter Fields | 72 |
| | SIP Headers Fields | 73 |
| | SIP Header, URI Component Fields | 75 |
| | SIP Header, URI Conversion Fields | 76 |
| | SIP Header, URI Parameter Fields | 78 |
| | SIP Header, Header Parameter Fields | 80 |
| | Adding a Normalization Policy | 83 |
| | Working With URI Components for a Request URI | 84 |

Working With URI Conversion Parameters for a Request URI 85

Working With URI Parameters for a Request URI 85

Working With SIP Headers 87

Working With URI Components for SIP Headers 88

Working With URI Conversion Parameters for SIP Headers 89

Working With URI Parameters for SIP Headers 90

Working With Header Parameters for SIP Headers 91

CHAPTER 11

Configuring Time Policies 93

Viewing a List of Time Policies 93

 About Time Policies 94

 Time Policy Fields 94

Adding a Time Policy 94

Viewing a List of Time Policy Steps 95

Adding or Editing a Time Policy Step 95

 Time Policy Steps 96

CHAPTER 12

Configuring Routing Triggers 99

Viewing a List of Routing Triggers 99

Adding or Editing a Routing Trigger 100

CHAPTER 13

Configuring Normalization Triggers 101

Viewing a List of Pre-Normalization Triggers 101

 About Normalization Triggers 101

Viewing a List of Post-Normalization Triggers 102

Adding and Editing a Pre-Normalization Trigger 102

Adding and Editing a Post-Normalization Trigger 103

CHAPTER 14

Configuring Performance Control 105

Configuring Performance Control 105

CHAPTER 15

Configuring Call Admission Control 107

Configuring Call Admission Control 107

| | | |
|-------------------|--|------------|
| CHAPTER 16 | Configuring Users | 109 |
| | Viewing a List of Users | 109 |
| | User Profile Fields | 110 |
| | Adding a New User | 111 |
| | Displaying or Changing a User Profile | 112 |
| | Displaying or Changing Group Subscriptions | 112 |
| | Finding a User | 113 |
| | Changing Your Password | 114 |

| | | |
|-------------------|------------------------------------|------------|
| CHAPTER 17 | Setting User Defaults | 115 |
| | User Defaults | 115 |
| | Configuring Password Options | 115 |
| | Configuring Account Lockout Policy | 116 |

| | | |
|-------------------|--|------------|
| CHAPTER 18 | Configuring Groups | 117 |
| | Viewing a List of Groups | 117 |
| | Group Fields | 118 |
| | Adding a New User Group | 118 |
| | Subscribing Members or Owners to a Group | 119 |
| | Unsubscribing Members and Owners from a Group | 120 |
| | Displaying or Modifying Group Parameters | 121 |
| | Viewing Owners and Members of a Group | 122 |
| | Modifying Group Ownership and Membership in Other Groups | 122 |
| | Deleting a Group | 123 |
| | Finding a Group | 124 |
| | About Capabilities | 124 |

| | | |
|-------------------|-------------------------------|------------|
| CHAPTER 19 | Configuring Privileges | 127 |
| | Viewing Privileges | 127 |
| | Overview of Privileges | 128 |
| | List of Operations | 128 |
| | Creating a Privilege | 130 |
| | Editing a Privilege | 130 |

| | |
|-------------------|--|
| CHAPTER 20 | Configuring Authentication, Authorization, and Accounting 133 |
| | Configuring the AAA Authentication Server 133 |
| | About the Authentication Order 133 |
| | About Authentication Failover 134 |
| | About Unreachable Failover 134 |
| | Example of Authentication Sequence 135 |
| | Configuring Connection Parameters for the AAA Authentication Server 135 |
| | Specifying the Policy that Controls the Behavior of Authentication and Authorization 136 |
| | Configure AAA Accounting Server 136 |
| | AAA Accounting Event Logging 137 |
| | Configuring the AAA Accounting Server 138 |
| | Configuring Accounting Event Logging 139 |
| | Configuring the AAA Accounting Server and Event Logging 139 |

| | |
|-------------------|---------------------------------------|
| CHAPTER 21 | Viewing System Information 141 |
| | System Information 141 |

| | |
|-------------------|---|
| CHAPTER 22 | Configuring Domain Name Settings 143 |
| | Changing a DNS Server 143 |
| | Adding a DNS Server 144 |
| | Removing a DNS Server 145 |

| | |
|-------------------|--|
| CHAPTER 23 | Configuring Network Time and Time Zone Settings 147 |
| | Adding NTP Server and Configuring Time Zone 147 |
| | Adding an NTP Server 147 |
| | Removing an NTP Server 148 |
| | Setting an NTP Server as the Preferred Server 149 |
| | Changing the Time Zone 149 |

| | |
|-------------------|--|
| CHAPTER 24 | Configuring SNMP Settings 151 |
| | About SNMP 151 |
| | Adding, Editing, and Deleting an SNMP Community String 151 |

| | | |
|-------------------|--|------------|
| | Adding, Editing, and Removing an SNMP Trap Host | 152 |
| | Enabling SNMP Traps | 153 |
| | Displaying MIBs | 153 |
| | Editing the SNMPv2-MIB | 154 |
| <hr/> | | |
| CHAPTER 25 | Configuring System Login Banner | 155 |
| <hr/> | | |
| CHAPTER 26 | Monitoring the Cisco Unified SIP Proxy System | 157 |
| | Monitoring the Call Statistics | 157 |
| | Monitoring the Server Group Status | 158 |
| | Server Group Status Page | 159 |
| <hr/> | | |
| CHAPTER 27 | Viewing Reports | 161 |
| | Viewing the Backup History Report | 161 |
| | Viewing the Restore History Report | 162 |
| | Viewing the Network Time Protocol Report | 162 |
| <hr/> | | |
| CHAPTER 28 | Configuring Backup and Restore | 165 |
| | Configuring the Backup Server | 165 |
| | Backup Configuration Parameters | 165 |
| | Viewing Scheduled Backups | 167 |
| | Adding a Scheduled Backup | 167 |
| | Manually Starting a Backup | 169 |
| | Starting a Restore | 169 |
| <hr/> | | |
| CHAPTER 29 | Using the Administration Control Panel | 171 |
| | Reloading Cisco Unified SIP Proxy | 171 |
| <hr/> | | |
| CHAPTER 30 | Managing the System Configuration | 173 |
| | Restoring System Defaults | 173 |
| | Viewing the Configuration Results | 174 |
| | Previewing the Candidate Configuration | 174 |

| | | |
|-------------------|-----------------------------------|------------|
| CHAPTER 31 | About Smart Licensing | 175 |
| | Configuring Smart License | 175 |
| | Smart Agent License Fields | 176 |
| | Viewing the Smart License Summary | 176 |

| | | |
|-------------------|----------------------------------|------------|
| CHAPTER 32 | Manage Inactivity Timeout | 179 |
| | Managing Inactivity Timeout | 179 |

| | | |
|-------------------|-------------------------------|------------|
| CHAPTER 33 | Patch Upgrade | 181 |
| | Downloading the Patch File | 181 |
| | Configuring Patch Upgrade | 182 |
| | Installing the Patch File | 182 |
| | Verifying Patch Upgrade | 183 |
| | Troubleshooting Patch Upgrade | 184 |

| | | |
|-------------------|--|------------|
| CHAPTER 34 | Troubleshooting | 185 |
| | Enabling Cisco Unified SIP Proxy Traces | 185 |
| | Component Levels | 186 |
| | Viewing the Cisco Unified SIP Proxy Log File | 187 |
| | Configuring Trace Settings | 187 |
| | Viewing Tech Support Information | 188 |
| | Viewing a Trace Buffer | 189 |
| | Viewing a Log File | 189 |
| | Enabling SIP Message Logging | 190 |
| | Searching SIP Message Calls | 191 |
| | Data for Call Search | 191 |
| | Viewing SIP Message Calls | 192 |
| | Enabling the Failed Calls Log | 193 |
| | Viewing the Failed Calls Log | 194 |
| | Viewing the History of a Failed Call | 194 |

| | | |
|-------------------|-----------------------|------------|
| CHAPTER 35 | Error Messages | 197 |
|-------------------|-----------------------|------------|

| | |
|------------------------------------|-----|
| CUSP Internal Error | 197 |
| Request Not Found | 197 |
| Authorization Failure | 197 |
| Configuration Prerequisite Missing | 198 |



CHAPTER 1

Change History

This table lists all the changes made to this guide, the most recent changes appearing at the top.

| Change | See | Date |
|---|--|----------------|
| Patch Upgrade | Patch Upgrade, on page 181 | April 30, 2020 |
| Updates to the minimum password length and password policy. | Changing Your Password, on page 114 | April 30, 2020 |
| SFTP Support | Backup Configuration Parameters, on page 165 | June 01, 2020 |



CHAPTER 2

Welcome to Cisco Unified SIP Proxy

- [Logging in to Cisco Unified SIP Proxy Graphical User Interface \(GUI\), on page 3](#)

Logging in to Cisco Unified SIP Proxy Graphical User Interface (GUI)



Note Mozilla Firefox is the recommended browser to use with the Cisco Unified SIP Proxy GUI.



Restriction Cisco Unified SIP Proxy restricts simultaneous administrator logins to the graphical user interface (GUI) .

Before you begin

- Install Cisco Unified SIP Proxy Release 10.2. See [Installation Guide for Cisco Unified SIP Proxy Release 10.2](#) for information.
- Gather the administrator username and password that you entered during installation.

SUMMARY STEPS

1. Open a web browser.
2. Enter the following URL: **https://<CUSP_IP_address>/admin/login**.
3. Enter the administrator name.
4. Enter the administrator password.
5. Click **Log In**.

DETAILED STEPS

Step 1 Open a web browser.

Step 2 Enter the following URL: **https://<CUSP_IP_address>/admin/login**.

The system displays the log-in screen.

Step 3 Enter the administrator name.

Step 4 Enter the administrator password.

Step 5 Click **Log In**.

The system displays the Cisco Unified SIP Proxy dashboard within the Cisco Unified SIP Proxy GUI.

About the Dashboard

The dashboard contains general information about the health and status of the system.

- Under the Server Group Status, the system displays the operational status of any server groups. The status can be either up or down.
- Under Call Routing Summary (Last Hour), the system displays the number of the following:
 - Total calls processed
 - Dropped calls
 - Peak CPS
 - Average CPS
 - Peak Supported CPS

Clicking on any of the first four headers takes you to the Monitoring page. See [Monitoring the Cisco Unified SIP Proxy System, on page 157](#). Clicking on the Peak Supported CPS header takes you to the Performance Control page. See [Configuring Performance Control, on page 105](#).

- Under Call Admission Control, the system displays the status of the call admission control feature. The feature can be either enabled or disabled. To enable or disable call admission control, see [Configuring Call Admission Control, on page 107](#).
- Under License Status, the system displays the number and the mode of license.

Commercial Open Source Licensing



CHAPTER 3

Configuring SIP Stacks

- [Viewing and Editing General Settings for SIP Stacks](#), on page 5
- [Adding and Deleting an Alias FQDN](#), on page 8
- [Adding and Deleting a Trusted Peer](#), on page 9

Viewing and Editing General Settings for SIP Stacks

SUMMARY STEPS

1. Choose **Configure** > **SIP Stack** > **General Settings**.
2. Update the values as described in the section [SIP Stack General Settings](#), on page 5.
3. Click **Update**.

DETAILED STEPS

Step 1 Choose **Configure** > **SIP Stack** > **General Settings**.

The system displays the SIP Stack Settings page with the SIP General Settings tab highlighted. It lists the general SIP settings.

Step 2 Update the values as described in the section [SIP Stack General Settings](#), on page 5.

Step 3 Click **Update**.

Related Topics

[SIP Stack General Settings](#), on page 5

SIP Stack General Settings

Table 1: SIP Stack General Settings

| Parameter | Description |
|-------------|-------------|
| SIP Message | |

| Parameter | Description |
|-----------------------|---|
| SIP Header Compaction | <p>Whether or not to enable SIP header compaction.</p> <p>When enabled, compact header forms are used for the following SIP headers:</p> <ul style="list-style-type: none"> • Call-ID • Contact • Content-Encoding • Content-Length • Content-Type • From • Subject • To • Via <p>When header compaction is disabled, complete SIP headers are used in all outgoing messages, regardless of the header format.</p> |
| SIP Message Logging | <p>Whether or not to enable the logging of all incoming and outgoing SIP messages.</p> <p>We recommend that you use the SIP Message Logging under SIP Stack General Settings to log messages without impacting the CUSP performance.</p> |
| SIP Statistics | Whether to display statistics for active SIP queues. |
| Period Time | (Optional, only available if you check Configuring SIP Stacks, on page 5) Determines how often to collect the peg-logging statistics. |
| Reset Time | (Optional, only available if you check Configuring SIP Stacks, on page 5) Determines how often to reset the peg-logging statistics. |

| Parameter | Description |
|---------------------|---|
| Max Forwards | <p>Specifies the maximum number of times that a request can be forwarded to another server. Each time a request is received by a server, this value is decremented by one. (If the request does not have a Max Forwards header, one is added.) When the value reaches zero, the server responds with a 483 (too many hops) response and terminates the transaction.</p> <p>You can use the Max Forwards header field to detect forwarding loops within a network.</p> <p>The allowed values are 0 to 255. The default value is 70.</p> <p>Note We recommend that you set this command to a value greater than or equal to 10, and less than or equal to 100.</p> |
| Overload | |
| Reject | Configures the server to send a 503 (Server Unavailable) response when the server is overloaded. |
| Retry After | <p>(Optional, only available if you choose Reject)</p> <p>The number of seconds sent in the SIP Retry-After header field of the 503 (Server Unavailable) response, which indicates when the sender can attempt the transaction again. If not specified, the 503 (Server Unavailable) response does not contain a Retry-After header field. The minimum value allowed is 0. The default value is 0.</p> |
| Redirect | Configures the server to send a 300 (Redirect) response when the server is overloaded. |
| IP Address | <p>(Optional, only available if you choose Redirect)</p> <p>The redirect interface host name or IP address sent in the SIP Contact header field. Subsequent requests will be redirected to the server at this address.</p> |
| Port | <p>(Optional, only available if you choose Redirect)</p> <p>The port of the redirect host. The valid range is from 1024 to 65535. The default is 5060.</p> |
| Transport Type | (Optional, only available if you choose Redirect) The transport protocol used by the redirect host. Can be UDP, TCP, or TLS. |
| DNS Settings | |
| DNS SRV Lookups | Configures SIP DNS SRV lookup commands. |

| Parameter | Description |
|-------------------------|---|
| DNS NAPTR Lookups | Enables the use of DNS NAPTR for domain hostname/IP address mapping. |
| TCP Settings | |
| Idle Connection Timeout | Configures the amount of idle time that is allowed to pass before sending a keep-alive probe. |
| Maximum Connections | Configures the maximum number of TCP/TLS connections. When the maximum number of TCP/TLS connections is reached, passive (incoming) connections are not accepted, and additional active (outgoing) connections can be made. |
| TLS Settings | |
| TLS Settings | Enables the use of SIP Transport Layer Security (TLS) connections with other SIP entities, providing secure communication over the Internet. TLS Versions 1.0, 1.1, and 1.2 can be enabled or disabled. |

Adding and Deleting an Alias FQDN

SUMMARY STEPS

1. Choose **Configure > SIP Stack > Alias FQDNs**.
2. To add an alias FQDN, do the following:
3. To delete an alias FQDN, do the following:

DETAILED STEPS

-
- Step 1** Choose **Configure > SIP Stack > Alias FQDNs**.
The system displays the Alias FQDNs page with the Alias FQDNs tab highlighted.
- Step 2** To add an alias FQDN, do the following:
- a) Enter a name.
 - b) Click **Add Alias**.
- Step 3** To delete an alias FQDN, do the following:
- a) Check the check box next to the name of the alias FQDN to delete.
 - b) Click **Remove**.
-

Adding and Deleting a Trusted Peer

This procedure creates one or more SIP TLS trusted peers. The establishment of TLS connections fails unless the identity of the remote side matches the identifier of a configured trusted peer. If there are no trusted peers configured, the connection is accepted as long as the TLS handshake succeeds.

SUMMARY STEPS

1. Choose **Configure > SIP Stack > TLS Trusted Peers**.
2. To add a TLS trusted peer, do the following:
3. To delete a TLS trusted peer, do the following:

DETAILED STEPS

Step 1 Choose **Configure > SIP Stack > TLS Trusted Peers**.

The system displays the TLS Trusted Peers page with the TLS Trusted Peers tab highlighted.

Step 2 To add a TLS trusted peer, do the following:

- a) Enter a name.
- b) Click **Add Trusted Peer**.

Step 3 To delete a TLS trusted peer, do the following:

- a) Check the check box next to the name of the TLS trusted peer to delete.
 - b) Click **Remove**.
-



CHAPTER 4

Configuring Networks

- [Viewing a List of Networks, on page 11](#)
- [Adding a Network, on page 11](#)
- [Editing the General Settings for a Network, on page 13](#)
- [Editing the SIP Retransmission Settings for a Network, on page 14](#)
- [Viewing and Deleting SIP Listen Points, on page 16](#)
- [Adding a SIP Listen Point, on page 16](#)
- [Editing the SIP Record-Route for a Network, on page 17](#)

Viewing a List of Networks

A SIP network is a logical collection of local interfaces that can be treated the same for general routing purposes.

Choose **Configure > Networks**.

The system displays the Networks page, listing all of the current networks.

Adding a Network



Restriction After a SIP network is created, you cannot remove it.

SUMMARY STEPS

1. Choose **Configure > Networks**.
2. Click **Add**.
3. Enter the information for the network as shown in the section [Network Information, on page 12](#).
4. Click **Add**.
5. To add a SIP Listen Point, do the following:

6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Networks**.
The system displays the Networks page.
- Step 2** Click **Add**.
The system displays the Network page.
- Step 3** Enter the information for the network as shown in the section [Network Information, on page 12](#).
- Step 4** Click **Add**.
The system displays the Networks page with all the networks listed, including the network that you just added.
- Step 5** To add a SIP Listen Point, do the following:
- Under the SIP Listen Points heading, click **click here** on the line for your network.
 - Click **Add**.
 - Enter the following required values:
 - IP address for the SIP Listen Point
 - Port for the SIP Listen Point
 - Transport type (UDP, TCP, or TLS) for the SIP Listen Point
 - Click **Add**.
- Step 6** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Network Information](#), on page 12

Network Information

Network Information

| Parameter | Description |
|-----------|---|
| Name | <p>Contains a name for this network. Network names can contain alphanumeric characters, period, dash, and underscore.</p> <p>Tip You cannot rename networks, so choose the network name carefully.</p> |

| Parameter | Description |
|---|---|
| Type | <p>Can be one of the following:</p> <ul style="list-style-type: none"> • standard—Configures the network interface to use standard SIP. The network has full UDP support. The network interface supports ICMP and different sockets can be used for each endpoint. • icmp—Configures the network interface to use Internet Control Message Protocol (ICMP). • noicmp—Specifies that the network interface does not use a separate socket for each endpoint. With this configuration, no ICMP errors are supported. • nat—Configures the network interface to use Network Address Translation (NAT). |
| Allow Outbound Connections | <p>Determines if you will allow this network to enable or disable outbound TCP/TLS client connections.</p> <p>Can be either enable or disable. Default value is enable.</p> |
| SIP Header Hiding: Hide VIA | <p>Check this check box to have the system strip the VIA header, so that downstream elements will not know the message path.</p> |
| UDP Settings: Maximum Packet Size | <p>Configures the maximum size of a UDP datagram for this network. The value must be between 1500 and 16,000.</p> |
| TCP Settings: TCP Connection Setup Timeout (ms) | <p>Configures the time (ms) up to which the TCP connection request waits before dropping the TCP connection request.</p> |
| TLS Certificate verification Setting | |
| Verify Client Certificate | <p>Enables client authentication verification for TLS connections.</p> |
| Verify Server Certificate | <p>Enables server authentication verification for TLS connections.</p> |

Editing the General Settings for a Network

Before you begin

You cannot edit the name of a network.

SUMMARY STEPS

1. Choose **Configure** > **Networks**.
2. Click the underlined name of the network.
3. Click the **General Settings** tab.
4. Update the values.
5. Click **Update**.
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure** > **Networks**.
The system displays the Networks page.
- Step 2** Click the underlined name of the network.
The system displays the Network '<name of the network>' page, with the information for the network. There are four tabs at the top of the page: General Settings, SIP Retransmissions, SIP Listen Points, and SIP Record-Route.
- Step 3** Click the **General Settings** tab.
- Step 4** Update the values.
- Step 5** Click **Update**.
- Step 6** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Editing the SIP Retransmission Settings for a Network

SUMMARY STEPS

1. Choose **Configure** > **Networks**.
2. Click the underlined name of the network.
3. Click the **SIP Retransmissions** tab. For more information, see [SIP Retransmissions, on page 15](#).
4. Update the values.
5. Click **Update**.
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure** > **Networks**.
The system displays the Networks page.
- Step 2** Click the underlined name of the network.
The system displays the Network '<name of the network>' page, with the information for the network.
- Step 3** Click the **SIP Retransmissions** tab. For more information, see [SIP Retransmissions, on page 15](#).

The system automatically populates many of the SIP retransmissions and timer fields.

Step 4 Update the values.

Step 5 Click **Update**.

Step 6 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[SIP Retransmissions](#), on page 15

SIP Retransmissions

Table 2: SIP Retransmissions

| Field | Description |
|---------------------------|--|
| T1 | Sets the initial request retransmission interval. |
| T2 | Sets the maximum request retransmission value. |
| T4 | Sets the amount of time a NONINVITE client transaction or INVITE server transaction remains active after completion to handle request or response retransmissions. |
| TU1 | Sets the amount of time an INVITE transaction remains active after completion with a 2xx response to handle response retransmissions. |
| TU2 | Sets the amount of time the server waits for a provisional or final response for an INVITE client transaction or NONINVITE server transaction after which the transaction is considered timed out. |
| clientTn | Sets the maximum lifetime of a client transaction. |
| serverTn | Sets the maximum lifetime of a server transaction. |
| Provisional (TU3) | (Optional) Configures SIP networks with TU3 transmission type only. |
| INVITE Client Transaction | Specifies the retransmit count for the INVITE request. |
| INVITE Server Transaction | Specifies the retransmit counts for final responses of INVITE requests. |
| Client Transaction | Specifies the retransmit count for requests other than INVITE. |

Viewing and Deleting SIP Listen Points

A SIP listen point, or listener, listens for SIP traffic on a specific SIP network, host, and port. You can configure multiple SIP listen points for a single network; however, you must create at least one before the server can accept SIP traffic.

- You do not have to disable listeners on the network when you make configuration changes to the network.
- You cannot run TCP and TLS listeners on the same port.

SUMMARY STEPS

1. Choose **Configure > Networks**.
2. To see the SIP listen points associated with a network, under the SIP Listen Points header, click **click here**.
3. To delete a SIP listen point, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Networks**.

The system displays the Networks page, listing all of the current networks.

Step 2 To see the SIP listen points associated with a network, under the SIP Listen Points header, click **click here**.

The system displays the Network '<name of the network>' page with the SIP Listen Points tab highlighted.

Note To see a different number of SIP listen points on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all SIP listen points. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.

Step 3 To delete a SIP listen point, do the following:

- a) Check the check box next to the name of the SIP listen point.
 - b) Click **Remove**.
-

Adding a SIP Listen Point

SUMMARY STEPS

1. Choose **Configure > Networks**.
2. To see the SIP listen points associated with a network, under the SIP Listen Points header, click **click here**.
3. To add a SIP listen point, do the following:
4. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Networks**.
The system displays the Networks page, listing all of the current networks.
- Step 2** To see the SIP listen points associated with a network, under the SIP Listen Points header, click **click here**.
The system displays the Network ‘<name of the network>’ page with the SIP Listen Points tab highlighted.
- Step 3** To add a SIP listen point, do the following:
- Click **Add**.
 - Enter the IP address, port, and transport type for the SIP listen point.
 - Click **Add**.
- Step 4** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Editing the SIP Record-Route for a Network

Before you begin

If your system is enabled for Lite Mode, then the system deletes the record route configurations and you cannot access the SIP Record-Route tab. To enable or disable Lite Mode, see [Configuring Performance Control, on page 105](#).

SUMMARY STEPS

- Choose **Configure > Networks**.
- Click the underlined name of the network.
- Click the **SIP Record-Route** tab.
- Choose either enable or disable.
- If you chose enable, enter the following information:
- Click **Update**.
- In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Networks**.
The system displays the Networks page.
- Step 2** Click the underlined name of the network.
The system displays the Network ‘<name of the network>’ page with the information for the network.
- Step 3** Click the **SIP Record-Route** tab.
- Step 4** Choose either enable or disable.
- Step 5** If you chose enable, enter the following information:

- Host for the SIP Record-Route
- Port for the SIP Record-Route
- Transport type (udp, tcp, or tls) for the SIP Record-Route

Step 6 Click **Update**.

Step 7 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173



CHAPTER 5

Configuring Triggers

- [Viewing and Deleting Triggers](#), on page 19
- [Adding a Trigger](#), on page 26
- [Viewing, Adding, Moving, and Deleting Rules for a Trigger](#), on page 27
- [Adding, Editing, and Deleting Conditions for a Trigger Rule](#), on page 28

Viewing and Deleting Triggers

SUMMARY STEPS

1. Choose **Configure** > **Triggers**.
2. To view the condition cases associated with this trigger, click the underlined name of the trigger.
3. To delete a trigger, do the following:

DETAILED STEPS

Step 1 Choose **Configure** > **Triggers**.

The system displays the Triggers page and displays all triggers.

Step 2 To view the condition cases associated with this trigger, click the underlined name of the trigger.

Step 3 To delete a trigger, do the following:

- a) Check the check box next to the name of the trigger to delete.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Configuring Triggers](#), on page 19

[Example of a Trigger](#), on page 20

[Available Trigger Conditions and Cases](#), on page 21

[Managing the System Configuration](#), on page 173

About Triggers

A trigger is a set of conditions that can be used to dictate routing and normalization logic. It is automatically executed in response to a certain event (or condition case). Conditions can have multiple cases.

Note the structure:

- A trigger is made up of one or more rules.
- A rule is made up of one or more conditions.
- A condition is made up of one or more cases.

For information on available triggers, see [Available Trigger Conditions and Cases, on page 21](#).

Example of a Trigger

You might have a trigger called `New_Trigger`. `New_Trigger` might have three rules, numbered 1, 2, and 3. Each rule has at least one condition and each condition has a case.

Table 3: Structure for the Trigger Called `New_Trigger`

| Trigger Rules | | | |
|---------------|-------|---|-----|
| | Logic | Condition | |
| 1 | | Inbound Network is exactly '100' | AND |
| | | Local IP Address is exactly '100.10.10.101' | AND |
| | | SIP Message request | |
| 2 | OR | Time Of Day is exactly '200' | AND |
| | | Mid-Dialog | AND |
| | | SIP Method UPDATE | |
| 3 | OR | Outbound Network is exactly '300' | AND |
| | | Transport Protocol tcp | |

In the previous table, the trigger is called `New_Trigger`. `New_Trigger` has three rules. Because of the “OR” logic, only one of the three rules has to be true before the trigger is launched.

Rule 1 has three conditions:

- Inbound Network is exactly '100'
- Local IP Address is exactly '100.10.10.101'
- SIP Message request

Because of the “AND” logic, all three conditions must be true before the rule is true.

In the condition “Inbound Network is exactly ‘100’”, the condition is “Inbound Network” and the case is “is exactly ‘100’”.

Available Trigger Conditions and Cases

The table lists the available trigger conditions and cases.

Table 4: Available Trigger Conditions and Cases

| Trigger Name | Trigger Description | Trigger Condition Case |
|------------------|--|---|
| Inbound Network | Configures the inbound network for a trigger condition for a server-side transaction. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |
| Outbound Network | Configures the outbound network for a trigger condition for a client-side transaction. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |

| Trigger Name | Trigger Description | Trigger Condition Case |
|-------------------|--|---|
| Local IP Address | Assigns a local-listen IP address that accepts incoming requests to a trigger condition. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |
| Local Port | Assigns a local-listen port to a trigger condition. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |
| Remote IP Address | Configures the remote IP network for a trigger condition. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |

| Trigger Name | Trigger Description | Trigger Condition Case |
|--------------|---|--|
| Remote Port | Configures the remote port for a trigger condition. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |
| SIP Message | Determines whether the trigger condition will fire based on whether the headers in the SIP message are request or response headers. | Enter the case: <ul style="list-style-type: none"> • request (default) • response |
| SIP Method | Configures a trigger condition in which the trigger is fired on the given SIP method name in the request. | <ul style="list-style-type: none"> • INVITE (default) • ACK • PRACK • UPDATE • BYE • REFER • INFO • MESSAGE • OPTIONS • SUBSCRIBE • NOTIFY • REGISTER • PUBLISH • regular expression |

| Trigger Name | Trigger Description | Trigger Condition Case |
|-------------------|--|---|
| SIP Response Code | Configures a trigger condition to fire on a specific response. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |
| SIP Header | Configures the trigger to fire when matching the regular expression for this header. | Set the SIP header name. Choose the SIP header index: <ul style="list-style-type: none"> • first (default) • last • all Choose the type of match: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex |
| Mid-Dialog | Configures the trigger to fire on mid-dialog responses. | none |
| Time Of Day | Configures the trigger to fire if the specified time policy is met. | Enter the case: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |

| Trigger Name | Trigger Description | Trigger Condition Case |
|--------------------|--|--|
| Transport Protocol | Assigns a transport protocol to the trigger condition. | Enter the case: <ul style="list-style-type: none"> • none (default) • udp • tcp • tls |
| Proxy Route | Ability to configure proxy route rule. | Choose the parameter: <ul style="list-style-type: none"> • uri (default) • uri-user • uri-host • uri-port • uri-scheme • uri-parameter • header-parameter Choose the type of match: <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex Enter the condition: <ul style="list-style-type: none"> • IP for remote IP address |

| Trigger Name | Trigger Description | Trigger Condition Case |
|--------------|--|---|
| Request URI | Configures a trigger to fire when matching the regular expression for the specified Uniform Resource Identifier (URI) parameter. | <p>Choose the parameter:</p> <ul style="list-style-type: none"> • uri (default) • uri-user • uri-host • uri-port • uri-scheme • uri-parameter • header-parameter <p>Choose the type of match:</p> <ul style="list-style-type: none"> • is exactly (default) • contains • starts with • ends with • regex <p>Enter the condition:</p> <ul style="list-style-type: none"> • IP for remote IP address |

Adding a Trigger



Restriction You cannot change the name of an existing trigger, so choose the name carefully.

SUMMARY STEPS

1. Choose **Configure > Triggers**.
2. Click **Add**.
3. Enter a name for this trigger.
4. To have only one rule apply before the trigger is activated (that is, to apply “OR” logic), add logic to the rule by checking the Logic box.
5. Click **Add**.
6. Add rules to the trigger. See [Viewing, Adding, Moving, and Deleting Rules for a Trigger, on page 27](#).
7. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Triggers**.
The system displays the Triggers page.
- Step 2** Click **Add**.
The system displays the Trigger (New) page.
- Step 3** Enter a name for this trigger.
- Step 4** To have only one rule apply before the trigger is activated (that is, to apply “OR” logic), add logic to the rule by checking the Logic box.
- Step 5** Click **Add**.
The system displays the Trigger ‘<name of the trigger>’ Conditions page.
- Step 6** Add rules to the trigger. See [Viewing, Adding, Moving, and Deleting Rules for a Trigger, on page 27](#).
- Step 7** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173

Viewing, Adding, Moving, and Deleting Rules for a Trigger

Before you begin

Add a trigger. See [Adding a Trigger, on page 26](#).

SUMMARY STEPS

1. Choose **Configure > Triggers**.
2. To view the rules for a trigger, click the underlined name of the trigger.
3. To add a rule for a trigger, do the following:
4. To delete a rule for a trigger, do the following:
5. If your trigger has multiple rules, you can reorder them by doing the following:
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Triggers**.
The system displays the Triggers page.
- Step 2** To view the rules for a trigger, click the underlined name of the trigger.
The system displays the Trigger ‘<name of the trigger>’ Rules page.
- Step 3** To add a rule for a trigger, do the following:

- a) Click **Add**. The system displays the Trigger '<name of the trigger>' Conditions page.
- b) Add conditions. See [Adding, Editing, and Deleting Conditions for a Trigger Rule, on page 28](#).

Step 4 To delete a rule for a trigger, do the following:

- a) Check the check box next to the rule to delete.
- b) Click **Remove**.

Step 5 If your trigger has multiple rules, you can reorder them by doing the following:

Tip The trigger fires as soon as a rule is matched. To optimize the system, we recommend that you put the rule most likely to match at the top of the list.

- a) Select the rule.
- b) Click the up or down arrows.
- c) Click **Update**.

Step 6 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration, on page 173](#)

Adding, Editing, and Deleting Conditions for a Trigger Rule



Note

- You cannot add condition cases to existing rules. You can only add condition cases to a rule when you originally create the rule.
 - You cannot edit existing conditions attached to a rule.
 - You cannot delete a condition case from a rule.
-

Before you begin

- Add a trigger and rules for the trigger. See [Adding a Trigger, on page 26](#) and [Viewing, Adding, Moving, and Deleting Rules for a Trigger, on page 27](#).

SUMMARY STEPS

1. Choose **Configure > Triggers**.
2. Click the underlined name of the trigger.
3. To add a rule, click **Add**.
4. To add a condition, do the following:
5. Add additional conditions to this rule as needed.

DETAILED STEPS

Step 1 Choose **Configure > Triggers**.

The system displays the Triggers page.

Step 2 Click the underlined name of the trigger.

The system displays the Trigger '<name of the trigger>' Rules page.

Step 3 To add a rule, click **Add**.

The system displays the Trigger '<name of the trigger>' Conditions page. You are automatically adding a new rule by being on this page. This page is where you add conditions to the new rule.

Step 4 To add a condition, do the following:

- a) Select a condition from the Trigger Condition drop-down menu. See [Available Trigger Conditions and Cases, on page 21](#).
- b) If necessary, select a condition case.
- c) If necessary, enter a condition to match.
- d) Click **Add**.

The system displays the Trigger '<name of the trigger>' Conditions page with the new condition.

Step 5 Add additional conditions to this rule as needed.

Related Topics

[Managing the System Configuration](#), on page 173

[Available Trigger Conditions and Cases](#), on page 21



CHAPTER 6

Configuring Server Groups

- [Viewing a List of Server Groups, on page 31](#)
- [Adding a Server Group, on page 33](#)
- [Editing a Server Group, on page 34](#)
- [Viewing and Editing the General Settings for All Server Groups, on page 35](#)
- [Viewing and Deleting Server Group Elements, on page 37](#)
- [Adding and Editing a Server Group Element, on page 39](#)
- [Viewing a List of SIP Ping Network Connections, on page 39](#)
- [Adding a SIP Ping Configuration, on page 41](#)
- [Editing a SIP Ping Configuration, on page 42](#)
- [Viewing a List of Call Admission Control Endpoints, on page 43](#)
- [Changing the Limit of a Call Admission Control Endpoint, on page 43](#)

Viewing a List of Server Groups

Server groups define the elements with which the Cisco Unified SIP Proxy system interacts for each network.

SUMMARY STEPS

1. Choose **Configure > Server Groups > Groups**.
2. To delete a server group, do the following:
3. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Server Groups > Groups**.

The system displays the Server Groups page with the Groups tab highlighted, containing the fields described in [Server Groups \(Group Tab\) Fields, on page 32](#).

Step 2 To delete a server group, do the following:

- a) Check the check box next to the server group to delete.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

- Step 3** To revert any changes you have made back to the state they were in at the time of the last commit, do the following:
- Check the check box next to the name of the server group that has the changes to which you want to revert.
 - Click **Revert**.
 - In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

- [Managing the System Configuration](#), on page 173
- [Server Groups \(Group Tab\) Fields](#), on page 32

Server Groups (Group Tab) Fields

Table 5: Server Groups (Groups Tab) Fields

| Parameter | Description |
|-----------|--|
| State | <p>Can be one of the following:</p> <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | <p>Name of this server group.</p> <p>Note The system inserts the server group name into the SIP URI of the outgoing request. Some devices, such as Cisco Unified Communications Manager, validate the URI of requests before processing, so you may need to configure the end device with a Fully Qualified Domain Name (FQDN) to allow for this functionality.</p> |

| Parameter | Description |
|-------------------------|--|
| Load Balancing Scheme | <p>Configures the load-balancing algorithm for all SIP server groups.</p> <p>Can be one of the following:</p> <ul style="list-style-type: none"> • global (default) • call-id—Specifies that a hash algorithm with call-id is performed to select an element. • request-uri—Specifies that a hash algorithm with a request URI is performed to select an element. • to-uri—Specifies that a hash algorithm with a To header URI is performed to select an element. • weight—Specifies that the element is selected proportional to its weight relative to the weights of other elements of the same q-value. This value is only applicable if implementing weight-based routing. • highest-q—Specifies that the first element in the list of available elements with the same highest q-value is selected. |
| Network | Name of the network associated with this server group. |
| Elements | Elements associated with this server group. |
| Pinging Allowed | Whether pinging is allowed. Can be either true or false. |
| Failover Response Codes | <p>The response code(s) that indicates the next-hop server is unable to process the request. The valid values are numbers between 500 and 599.</p> <p>To add multiple failover response codes, separate the individual codes by a comma and indicate ranges with a dash. Commas and dashes must be followed by a space.</p> |

Adding a Server Group

Before you begin

You must create and configure at least one network before you can add a server group. See [Configuring Networks](#), on page 11.

SUMMARY STEPS

1. Choose **Configure > Server Groups > Groups**.
2. Click **Add**.
3. Enter information. See [Server Groups \(Group Tab\) Fields, on page 32](#).
4. Click **Add**.
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Server Groups > Groups**.
The system displays the Server Groups page with the Groups tab highlighted.
- Step 2** Click **Add**.
The system displays the Server Group (New) page.
- Step 3** Enter information. See [Server Groups \(Group Tab\) Fields, on page 32](#).
- Step 4** Click **Add**.
- Step 5** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

- [Managing the System Configuration, on page 173](#)
- [Configuring Server Groups, on page 31](#)
- [Server Groups \(Group Tab\) Fields, on page 32](#)

Editing a Server Group

SUMMARY STEPS

1. Choose **Configure > Server Groups > Groups**.
2. Click the underlined name of the server group to edit.
3. Edit the information. See [Server Groups \(Group Tab\) Fields, on page 32](#).
4. Click **Update**.
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Server Groups > Groups**.
The system displays the Server Groups page with the Groups tab highlighted.
- Step 2** Click the underlined name of the server group to edit.
The system displays the Server Group '<name of server group>' page with the Group Settings tab highlighted.
- Step 3** Edit the information. See [Server Groups \(Group Tab\) Fields, on page 32](#).

Step 4 Click **Update**.

Step 5 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

[Server Groups \(Group Tab\) Fields](#), on page 32

Viewing and Editing the General Settings for All Server Groups

Follow this procedure to view and edit the general settings that affect all server groups.

SUMMARY STEPS

1. Choose **Configure > Server Groups > General Settings**.
2. To edit the settings, change the values.
3. Click **Update**.
4. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure > Server Groups > General Settings**.

The system displays the Server Groups page with the General Settings tab highlighted, containing the fields described in [Server Groups \(General Settings Tab\) Fields, on page 35](#).

Step 2 To edit the settings, change the values.

Step 3 Click **Update**.

Step 4 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

[Server Groups \(General Settings Tab\) Fields](#), on page 35

Server Groups (General Settings Tab) Fields

Table 6: Server Groups (General Settings Tab) Fields

| Parameter | Description |
|------------------------------|-------------|
| Server Group Element Retries | |

| Parameter | Description |
|--|--|
| UDP | Maximum number of consecutive failed attempts to send a request to a server group element via the specified protocol before the element is considered down. A failed attempt can occur because of a timeout, ICMP error, or receipt of a failure response. The valid range is from 0 to 65535. |
| TCP | |
| TLS | |
| Global Load Balancing Scheme | |
| Load Balancing Scheme | <p>Configures the load-balancing algorithm for all SIP server groups.</p> <p>Can be one of the following:</p> <ul style="list-style-type: none"> • call-id (default)—Specifies that a hash algorithm with call-id is performed to select an element. • request-uri—Specifies that a hash algorithm with a request URI is performed to select an element. • to-uri—Specifies that a hash algorithm with a To header URI is performed to select an element. • weight—Specifies that the element is selected proportional to its weight relative to the weights of other elements of the same q-value. This value is only applicable if implementing weight-based routing. • highest-q—Specifies that the first element in the list of available elements with the same highest q-value is selected. |
| Global Ping | |
| Pinging Allowed | Whether pinging is allowed. Can be either enable or disable. |
| Ping 503 | |
| Verifies the 503 response code | Checks whether the SIP application service in the remote server element is up or down by monitoring the response. It treats the element as down for the 503 response to the PING request. |
| Default Failed Element Retry After Duration (in milliseconds) | |

| Parameter | Description |
|-------------------------|---|
| Failover Response Codes | <p>The response code(s) that indicates the next-hop server is unable to process the request. The valid values are numbers between 500 and 599.</p> <p>To add multiple failover response codes, separate the individual codes by a comma and indicate ranges with a dash. Commas and dashes must be followed by a space.</p> |

Viewing and Deleting Server Group Elements

There can be multiple elements in each server group.

SUMMARY STEPS

1. Choose **Configure > Server Groups > Groups**.
2. To see the elements associated with this server group, click **click here** under the Elements header.
3. To delete a server group element, do the following:
4. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Server Groups > Groups**.

The system displays the Server Groups page with the Groups tab highlighted.

Step 2 To see the elements associated with this server group, click **click here** under the Elements header.

The system displays the Server Group '<name of server group>' page with the Elements tab highlighted. The page contains the fields described in [Server Group \(Elements Tab\) Fields, on page 38](#).

Step 3 To delete a server group element, do the following:

- a) Check the check box next to the name of the element.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Step 4 To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

- a) Check the check box next to the name of the server group element that has the changes to which you want to revert.
- b) Click **Revert**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

[Server Group \(Elements Tab\) Fields](#), on page 38

Server Group (Elements Tab) Fields

Table 7: Server Group (Elements Tab) Fields

| Parameter | Description |
|---------------------|--|
| State | <p>Can be one of the following:</p> <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| IP Address | Specifies the interface host name or IP address of the server group element. |
| Port | Specifies the port used by the server group element. Valid values are from 1024 to 65535. The default is 5060. |
| Transport | <p>Specifies the transport type of the server group element. Can be one of the following:</p> <ul style="list-style-type: none"> • UDP (default) • TCP • TLS |
| Nested Server Group | Whether or not this server group can contain another server group. |
| Q-Value | <p>Specifies a real number that indicates the priority of the server group element with respect to others in the server group.</p> <p>The Q-value provides the priority of each member (element) which varies from 0.0 to 1.0, where 1.0 is the highest priority.</p> |
| Weight | <p>Specifies the percentage assigned to the IP element in the server group if implementing weight-based routing.</p> <p>The valid range is from 0 to 100. The default weight is 0.</p> |

Adding and Editing a Server Group Element

SUMMARY STEPS

1. Choose **Configure > Server Groups > Groups**.
2. Click **click here** under the Elements header that corresponds with the server group to which you want to add an element.
3. To add an element, do the following:
4. To edit an element, do the following:
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Server Groups > Groups**.
- The system displays the Server Groups page with the Groups tab highlighted.
- Step 2** Click **click here** under the Elements header that corresponds with the server group to which you want to add an element.
- The system displays the Server Group ‘<name of server group>’ page with the Elements tab highlighted.
- Step 3** To add an element, do the following:
- a) Click **Add**. The system displays the Server Group ‘<name of server group>’ Element (New) page.
 - b) Choose whether this element will be for an endpoint or server group.
 - c) Enter information about the element as described in [Server Group \(Elements Tab\) Fields, on page 38](#).
 - d) Click **Add**.
- Step 4** To edit an element, do the following:
- a) Click the underlined IP address for the element to edit. The system displays the Server Group ‘<name of server group>’ Element page.
 - b) Make changes to the values.
 - c) Click **Update**.
- Step 5** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

[Server Group \(Elements Tab\) Fields](#), on page 38

Viewing a List of SIP Ping Network Connections

Before you begin

You must have already created at least one network. See [Configuring Networks, on page 11](#).

SUMMARY STEPS

1. Choose **Configure > Server Groups > SIP Ping**.
2. To delete a SIP ping network connection, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Server Groups > SIP Ping**.

The system displays the SIP Ping page with the SIP Ping tab highlighted, containing the fields described in [SIP Ping Fields, on page 40](#).

Step 2 To delete a SIP ping network connection, do the following:

- a) Check the check box next to the SIP ping network connection to delete.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

[Configuring Server Groups](#), on page 31

[SIP Ping Fields](#), on page 40

SIP Ping Fields

Table 8: SIP Ping Fields

| Parameter | Description |
|------------|--|
| Network | Name of this SIP ping network connection. |
| IP Address | Specifies the interface host name or IP address that listens for responses to the SIP pings. Note When you specify a hostname, the server performs a DNS lookup to confirm that the host can be resolved. It then uses the IP address when the configuration is saved. If the system cannot resolve the hostname, it displays an “IP Address validation failed” error. |
| Port | The UDP port that listens for responses to the SIP pings. The valid range is from 1024 to 65535. The default value is 4000. Note Be sure this port number is different from the port number specified for the server’s SIP listen point. |

| Parameter | Description |
|---------------|--|
| SIP Method | The request method for the SIP pings. Can be one of the following: <ul style="list-style-type: none"> • OPTIONS (default) • PING • INFO |
| Ping Type | The ping type for the SIP ping. Can be one of the following: <ul style="list-style-type: none"> • Proactive—Specifies that pinging is performed to both up and down elements, and both are pinged at the same interval. • Reactive—Specifies that pinging is performed to only down elements. This is the default value. • Adaptive—Specifies that pinging is performed to both up and down elements, and both are pinged at different intervals. |
| Up Interval | (Optional; only available if you choose Adaptive or Proactive for Ping Type) Specifies the consecutive ping interval for up elements. The default value is 5000 milliseconds. |
| Down Interval | (Optional; only available if you choose Adaptive or Reactive for Ping Type) Specifies the consecutive ping interval in milliseconds. For Adaptive pinging, this value configures the down element ping interval. The default value is 5000 milliseconds. |
| Ping Timeout | Specifies the maximum number of milliseconds between a ping and a response before the ping is considered unsuccessful. The minimum allowed value is 0. The default value is 500. |

Adding a SIP Ping Configuration



Restriction

- You can only define one SIP ping configuration for each network. To create multiple SIP ping configurations, you must create and configure multiple networks.
- You can only add a SIP ping for server group elements with a transport type of UDP.

Before you begin

You must create and configure at least one network before you can add a SIP ping configuration. See [Configuring Networks, on page 11](#).

SUMMARY STEPS

1. Choose **Configure > Server Groups > SIP Ping**.
2. Click **Add**.
3. Enter information. See [SIP Ping Fields, on page 40](#).
4. Click **Add**.
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Server Groups > SIP Ping**.
The system displays the SIP Ping page with the SIP Ping tab highlighted.
- Step 2** Click **Add**.
The system displays the SIP Ping Configuration (New) page.
- Step 3** Enter information. See [SIP Ping Fields, on page 40](#).
- Step 4** Click **Add**.
- Step 5** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

- [Managing the System Configuration, on page 173](#)
- [SIP Ping Fields, on page 40](#)
- [Configuring Server Groups, on page 31](#)

Editing a SIP Ping Configuration

SUMMARY STEPS

1. Choose **Configure > Server Groups > SIP Ping**.
2. Check the check box next to the SIP ping network configuration to edit.
3. Click **Edit**.
4. Edit information. See [SIP Ping Fields, on page 40](#).
5. Click **Update**.
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Server Groups > SIP Ping**.

The system displays the SIP Ping page with the SIP Ping tab highlighted.

Step 2 Check the check box next to the SIP ping network configuration to edit.

Step 3 Click **Edit**.

The system displays the SIP Ping Configuration '<name of network>' page.

Step 4 Edit information. See [SIP Ping Fields](#), on page 40.

Step 5 Click **Update**.

Step 6 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

[SIP Ping Fields](#), on page 40

Viewing a List of Call Admission Control Endpoints

The system automatically adds call admission control endpoints when you add a server group and elements and then commit the configuration.

Choose **Configure > Server Groups > Call Admission Control**.

The system displays the Server Groups page with the Call Admission Control tab highlighted.

For each call admission control endpoint, the system lists the IP address, port, transport, network and call admission control limit.

Related Topics

[Managing the System Configuration](#), on page 173

Changing the Limit of a Call Admission Control Endpoint

Step 1 Choose **Configure > Server Groups > Call Admission Control**.

The system displays the Server Groups page with the Call Admission Control tab highlighted.

Step 2 Click the underlined limit to change.

The system displays the CAC Endpoint page.

Step 3 Check the **unlimited** check box to make the limit unlimited, or enter a value in the field.

Step 4 Click **Update**.

Related Topics

[Configuring Call Admission Control](#), on page 107



CHAPTER 7

Configuring Route Groups

- [Viewing a List of Route Groups and Corresponding Elements, on page 45](#)
- [Adding a Route Group, on page 48](#)
- [Viewing and Deleting Route Group Elements, on page 49](#)
- [Adding and Editing Route Group Elements, on page 50](#)
- [Editing a Route Group, on page 50](#)

Viewing a List of Route Groups and Corresponding Elements

SUMMARY STEPS

1. Choose **Configure > Route Groups**.
2. There can be multiple elements in a route group. To see the elements associated with this route group, click **click here**.
3. To delete a route group, do the following:
4. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

-
- Step 1** Choose **Configure > Route Groups**.
- The system displays the Route Groups page, which contains the fields described in [Route Group Fields, on page 46](#).
- Step 2** There can be multiple elements in a route group. To see the elements associated with this route group, click **click here**.
- The system displays the Route Group ‘<name of route group>’ page, containing the fields described in [Element Fields, on page 47](#).
- Step 3** To delete a route group, do the following:
- a) Check the check box next to the name of the route group to delete.
 - b) Click **Remove**.
 - c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
- Step 4** To revert any changes you have made back to the state they were in at the time of the last commit, do the following:
- a) Check the check box next to the name of the route group that has the changes to revert back to.

- b) Click **Revert**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

About Route Groups

A route group allows you to designate the order in which gateways and trunks are selected. It allows you to prioritize a list of gateways and ports for outgoing trunk selection.

For example, if you use two long-distance carriers, you could add a route group so that long-distance calls to the less expensive carrier are given priority. Calls only route to the more expensive carrier if the first trunk is unavailable.

You can add, update, or delete route groups from the Route Group page. You can also add, update, or delete elements.

Route Group Fields

The table lists the fields on the Route Groups page.

Table 9: Route Group Parameters

| Parameter | Description |
|----------------------|---|
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Name of this route group. |
| Elements | Elements that belong to this route group. |
| Time Of Day Routing | Specifies if this route group allows for time policy-based routing. Can be either true or false. The default value is false. |
| Weight Based Routing | Specifies if this route group allows for weight-based routing. Can be either true or false. The default value is false. |

Element Fields

The table lists the fields on the Route Group ‘<name of route group>’ page when the Elements tab is highlighted.

Table 10: Route Group Element Parameters

| Parameter | Description |
|---------------------------|--|
| State | <p>Can be one of the following:</p> <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Target Destination | |
| Host/Server Group | <p>Specifies the interface host name or IP address of the route group element.</p> <p>Note If you select Server Group, you must not enter the port and transport type details.</p> |
| Port | <p>Specifies the port used by the route group element. Valid values are from 1024 to 65535. The default is 5060.</p> |
| Transport | <p>Specifies the transport type of the route group element.</p> <p>Can be one of the following:</p> <ul style="list-style-type: none"> • none (default) • UDP • TCP • TLS |
| Next Hop | |
| SIP URI | The URI of the next hop. |
| Options | |
| Network | The name of the network to which this route group is associated. |

| Parameter | Description |
|-------------------------|--|
| Q-Value | (Optional) Specifies a real number that indicates the priority of the route group element with respect to others in the route group. The Q-value provides the priority of each member (element) which varies from 0.0 to 1.0, where 1.0 is the highest priority. |
| Weight | (Optional) Specifies the percentage assigned to the IP element in the route group if implementing weight-based routing. The valid range is from 0 to 100. The default weight is 0. |
| Time Policy | Specifies the time policy if time-based routing is being used. |
| Failover Response Codes | The response code(s) that indicates the next-hop server is unable to process the request. The valid values are numbers between 400 and 599. To add multiple failover response codes, separate the individual codes by a comma and indicate ranges with a dash. Commas and dashes must be followed by a space. |

Related Topics

[Managing the System Configuration](#), on page 173

Adding a Route Group

SUMMARY STEPS

1. Choose **Configure > Route Groups**.
2. Click **Add**.
3. Enter a name for this route group. If you will enable time-of-day routing or weight-based routing, check those check boxes.
4. Click **Add**.
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Route Groups**.
The system displays the Route Groups page.
- Step 2** Click **Add**.

The system displays the Route Group (New) page.

Step 3 Enter a name for this route group. If you will enable time-of-day routing or weight-based routing, check those check boxes.

Step 4 Click **Add**.

The system displays the Route Groups page, with the new route group listed in the table.

Step 5 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Viewing and Deleting Route Group Elements

SUMMARY STEPS

1. Choose **Configure > Route Groups**.
2. On the line of the route group that has the element to delete, under the title Elements, click **click here**.
3. To delete a route group element, do the following:
4. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Route Groups**.

The system displays the Route Groups page.

Step 2 On the line of the route group that has the element to delete, under the title Elements, click **click here**.

The system displays the Route Group '<name of route group>' page with the Elements tab highlighted.

Step 3 To delete a route group element, do the following:

- a) Check the check box next to the name of the element.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Step 4 To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

- a) Check the check box next to the name of the route group element that has the changes to revert back to.
- b) Click **Revert**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Adding and Editing Route Group Elements

SUMMARY STEPS

1. Choose **Configure > Route Groups**.
2. Under Elements, click **click here** on the line for the route group for which you want to add an element.
3. To add an element, do the following:
4. To edit an element, do the following:
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Route Groups**.
The system displays the Route Groups page.
- Step 2** Under Elements, click **click here** on the line for the route group for which you want to add an element.
The system displays the Route Group ‘<name of route group>’ page with the Elements tab highlighted.
- Step 3** To add an element, do the following:
- a) Click **Add**. The system displays the Route Group ‘<name of route group>’ Element (New) page.
 - b) Choose whether this element will use a target destination or next hop.
 - c) Enter information about the element as described in [Element Fields, on page 47](#).
 - d) Click **Add**.
- Step 4** To edit an element, do the following:
- a) Click the underlined Host/Server Group of the element. The system displays the Route Group ‘<name of route group>’ Element page.
 - b) Make changes to the information about the element as described in [Element Fields, on page 47](#).
 - c) Click **Update**.
- Step 5** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Editing a Route Group

SUMMARY STEPS

1. Choose **Configure > Route Groups**.
2. Click the underlined name of the route group to edit.
3. You can change if this route group will enable time-of-day routing or weight-based routing.
4. Click **Update**.

5. To edit the elements of the route group, follow the procedure [Adding and Editing Route Group Elements, on page 50](#).
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Route Groups**.
The system displays the Route Groups page.
- Step 2** Click the underlined name of the route group to edit.
The system displays the Route Group '<name of route group>' page with the Group Settings tab highlighted.
- Step 3** You can change if this route group will enable time-of-day routing or weight-based routing.
- Step 4** Click **Update**.
- Step 5** To edit the elements of the route group, follow the procedure [Adding and Editing Route Group Elements, on page 50](#).
- Step 6** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173



CHAPTER 8

Configuring Route Tables

- [Viewing a List of Route Tables](#), on page 53
- [Adding a Route Table](#), on page 56
- [Viewing a List of Route Table Routes](#), on page 57
- [Adding a Route to a Route Table](#), on page 58
- [Exporting Active Routes](#), on page 59
- [Editing the Routes Associated with a Route Table](#), on page 59

Viewing a List of Route Tables

SUMMARY STEPS

1. Choose **Configure > Route Tables**.
2. To delete a route table, do the following:
3. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

- Step 1** Choose **Configure > Route Tables**.
- The system displays the Route Tables page, containing the fields described in the section Route Table Fields.
- Step 2** To delete a route table, do the following:
- a) Check the check box next to the name of the route table to delete.
 - b) Click **Remove**.
 - c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
- Step 3** To revert any changes you have made back to the state they were in at the time of the last commit, do the following:
- a) Check the check box next to the name of the route table that has the changes to revert back to.
 - b) Click **Revert**.
 - c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173

About Route Tables

You configure route tables to direct SIP requests to their appropriate destinations. Each route table consists of a set of keys that are matched based on the lookup policy.

For example, in one table, each key might represent the prefix of the phone number dialed. The table performs a task depending on the prefix dialed. In this example, the table is designed to respond to calls with a 404 message (not found) unless the phone number dialed begins with 510. Another table might be designed to respond to calls with a 404 message (not found) unless the phone number dialed begins with the escape sequence (91).

You can add, update, or delete route tables from the Route Tables page. You can also add, update, or delete routes.

Related Topics

[Managing the System Configuration](#), on page 173

Route Table Fields

The table lists the fields on the Route Tables page.

Table 11: Route Tables Parameters

| Parameter | Description |
|-----------|---|
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Name of this route table. The valid characters are alphanumeric characters, dash, period, and underscore. |

Related Topics

[Viewing a List of Route Tables](#), on page 53

[Managing the System Configuration](#), on page 173

Route Fields

The table lists the fields on the Route Table '<name of route>' Routes page.



Note Depending on the route type that you choose, you will see some or all of these parameters.

Table 12: Route Table Route Parameters

| Parameter | Description |
|---|---|
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Candidate Value | |
| Key | Specifies the route table lookup key number. The lookup key represents the portion of the SIP message that is being matched, and must be unique to the routing table. |
| Route Type | Can be one of the following: <ul style="list-style-type: none"> • destination • route-group • route-policy • response • default-sip |
| Destination Route Type (Optional; only available if you choose a Route Type of destination or default-sip) | |
| Destination Route Type | The type of route. Can be either target destination, next hop, or both. |
| Network | Specifies the SIP network name. |
| Target Destination (Optional; only available if you choose a Destination Route Type of target destination or both) | |
| Host/Server Group | Hostname or IP address of the target destination. <p>Note If you select Server Group, you must not enter the port and transport type details.</p> |

| Parameter | Description |
|---|--|
| Port | Port of the target destination. Values can be 1024 to 65535. |
| Transport Type | Can be one of the following: <ul style="list-style-type: none"> • none • UDP • TCP • TLS |
| Next Hop (Optional; only available if you choose a Destination Route Type of next hop or both) | |
| SIP URI | URI of the next hop. |
| Route-Group Route Type (Optional; only available if you choose a Route Type of route-group) | |
| Route Group | The name of the route group. |
| Response Route Type (Optional; only available if you choose a Route Type of response) | |
| Response | Specifies the response code to a lookup key in a routing table. |
| Route-Policy Route Type (Optional; only available if you choose a Route Type of route-policy) | |
| Lookup Route Policy | Specifies the route lookup policy to be used in the routing table. |
| Default SIP Route | Simple routing following RFC 3263. |

Related Topics

[Viewing a List of Route Tables](#), on page 53

[Managing the System Configuration](#), on page 173

Adding a Route Table

SUMMARY STEPS

1. Choose **Configure > Route Tables**.
2. Click **Add**.
3. Enter a name for this route table.
4. Click **Add**.
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Route Tables**.
The system displays the Route Tables page.
- Step 2** Click **Add**.
The system displays the Route Tables page.
- Step 3** Enter a name for this route table.
- Step 4** Click **Add**.
The system displays the Route Tables page, with the new route table listed.
- Step 5** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173

Viewing a List of Route Table Routes

SUMMARY STEPS

1. Choose **Configure > Route Tables**.
2. To see the routes associated with the route table, click the underlined name of the route table.
3. To see a different number of routes on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all routes.
4. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number, and press **Enter**.
5. To delete a route, do the following:
6. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

- Step 1** Choose **Configure > Route Tables**.
The system displays the Route Tables page, containing the fields described in [Route Table Fields, on page 54](#).
- Step 2** To see the routes associated with the route table, click the underlined name of the route table.
The system displays the Route Table '<name of route table>' Routes page, containing some or all of the fields described in [Route Fields, on page 54](#).
- Step 3** To see a different number of routes on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all routes.
- Step 4** To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number, and press **Enter**.

- Step 5** To delete a route, do the following:
- Check the check box next to the name of the route to delete.
 - Click **Remove**.
 - In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
- Step 6** To revert any changes you have made back to the state they were in at the time of the last commit, do the following:
- Check the check box next to the name of the route table that has the changes to revert back to.
 - Click **Revert**.
 - In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Adding a Route to a Route Table

SUMMARY STEPS

- Choose **Configure > Route Tables**.
- Click the underlined name of the route table for which you want to add a route.
- Click **Add**.
- Enter information about the route as described in [Route Fields, on page 54](#).
- Click **Add**.
- In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Route Tables**.
- The system displays the Route Tables page.
- Step 2** Click the underlined name of the route table for which you want to add a route.
- The system displays the Route Table '<name of route table>' Routes page.
- Step 3** Click **Add**.
- The system displays the Route Table '<name of route table>' Route (New) page.
- Step 4** Enter information about the route as described in [Route Fields, on page 54](#).
- Step 5** Click **Add**.
- Step 6** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Exporting Active Routes



Restriction You can only export routes that are in the active state. To move a route to the active state, commit the configuration.

SUMMARY STEPS

1. Choose **Configure > Route Tables**.
2. Click the underlined name of the route table that contains the routes to export.
3. Click **Export Active Routes**.
4. Click **Save**.
5. Enter the location to which you want to export the file. Click **OK**.

DETAILED STEPS

-
- Step 1** Choose **Configure > Route Tables**.
The system displays the Route Tables page.
- Step 2** Click the underlined name of the route table that contains the routes to export.
The system displays the Route Table '<name of route table>' Routes page.
- Step 3** Click **Export Active Routes**.
The system displays a dialog box.
- Step 4** Click **Save**.
- Step 5** Enter the location to which you want to export the file. Click **OK**.
The system saves the route to that location.
-

Editing the Routes Associated with a Route Table

SUMMARY STEPS

1. Choose **Configure > Route Tables**.
2. Click the underlined name of the route table that contains the route to edit.
3. Click the underlined name of the key for the route to edit.
4. Make changes to the values.
5. Click **Update**.
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Route Tables**.
The system displays the Route Tables page.
- Step 2** Click the underlined name of the route table that contains the route to edit.
The system displays the Route Table '<name of route table>' Routes page.
- Step 3** Click the underlined name of the key for the route to edit.
The system displays the Route Table '<name of route table>' Route page.
- Step 4** Make changes to the values.
- Step 5** Click **Update**.
- Step 6** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173



CHAPTER 9

Configuring Route Policies

- [Viewing a List of Route Policies, on page 61](#)
- [Adding a Route Policy, on page 64](#)
- [Viewing a List of Route Policy Steps, on page 65](#)
- [Adding or Editing a Route Policy Step, on page 66](#)

Viewing a List of Route Policies

A route policy defines the behavior of the route.



Note Route policies are also called lookup policies in the CLI.

SUMMARY STEPS

1. Choose **Configure > Route Policies**.
2. To delete a route policy, do the following:
3. To revert a route policy to the settings it had at the time of the last commit, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Route Policies**.

The system displays the Route Policies page, containing the fields described in [Route Policy Fields, on page 62](#).

Step 2 To delete a route policy, do the following:

- a) Check the check box next to the name of the route policy to delete.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Step 3 To revert a route policy to the settings it had at the time of the last commit, do the following:

- a) Check the check box next to the name of the route policy whose settings you want to revert back to.
- b) Click **Revert**.

c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Route Policy Fields

The table lists the fields on the Route Policies page.

Table 13: Route Policy Fields

| Parameter | Description |
|-----------|---|
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Name of this route policy. |

Route Policy Step Fields

The table lists the fields on the Route Policy Step page.

Table 14: Route Policy Step Fields

| Parameter | Description |
|--------------------|---|
| Route Table | |
| Name | The name of the route table to which this route policy is attached. |

| Parameter | Description |
|-------------------------------|--|
| Lookup Key Matches: | <p>Can be one of the following:</p> <ul style="list-style-type: none"> • Exactly (default)—Specifies that the lookup policy searches for the exact match of the key in the specified table. • Prefix-Longest-Match—Specifies that the lookup policy searches for the longest prefix match. • Subdomain—Specifies that the lookup policy searches for the longest subdomain of the keys in the table. Domain name matching is case-sensitive and the most specific match prevails, and IP address matching must be exact. If a request contains a non-SIP request URI, this lookup fails. To prevent this from happening, check the check box next to Case Sensitive. • Subnet—Specifies that the lookup policy searches for the longest IP addresses of the keys in the table. • Prefix-Fixed-Length—Specifies that a fixed number of characters from the key is looked up instead of the complete key. |
| Case Sensitive | Check this check box if you want the lookup policy for the route table to be case sensitive. |
| Route Table Lookup Key | |

| Parameter | Description |
|-----------------------------|--|
| Lookup Key | <p>Select a target destination from the drop-down menu. Values are:</p> <ul style="list-style-type: none"> • Request URI—Specifies the lookup policy to apply to the Request-URI header. • Field • SIP Header—Specifies the header for which the lookup policy is applicable. <p>Select a URI component from the drop-down menu, Values are:</p> <ul style="list-style-type: none"> • URI—Specifies the lookup policy to apply to the full URI. • User—Specifies the lookup policy to apply to the user URI component. • Phone—Specifies the lookup policy to apply to the phone URI component. • Host—Specifies the lookup policy to apply to the host URI component. • Host-Port—Specifies the lookup policy to apply to the host-port URI component. • Param—Specifies the URI component parameter name. |
| Lookup Key Modifiers | |
| Regular Expression Match | Specifies the key modifier to match the regular expression. |
| Regular Expression Replace | Specifies the key modifier to replace the regular expression. |

Related Topics

[Managing the System Configuration](#), on page 173

Adding a Route Policy

Before you begin

You must create and configure at least one route table before you can add a route policy. See [Configuring Route Tables](#), on page 53.

SUMMARY STEPS

1. Choose **Configure > Route Policies**.
2. Click **Add**.
3. Enter a name for this route policy.
4. Enter route policy steps. See [Adding or Editing a Route Policy Step, on page 66](#).
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Route Policies**.
The system displays the Route Policies page.
- Step 2** Click **Add**.
The system displays the Route Policy (New) page.
- Step 3** Enter a name for this route policy.
Click **Add**.
The system displays the Route Policy Step (New) page.
- Step 4** Enter route policy steps. See [Adding or Editing a Route Policy Step, on page 66](#).
- Step 5** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

- [Configuring Route Policies, on page 61](#)
- [Managing the System Configuration, on page 173](#)

Viewing a List of Route Policy Steps

SUMMARY STEPS

1. Choose **Configure > Route Policies**.
2. Click the underlined name of the route policy for which you want to see the route policy steps.
3. To delete a route policy step, do the following:
4. To revert a route policy step to the settings it had at the time of the last commit, do the following:

DETAILED STEPS

-
- Step 1** Choose **Configure > Route Policies**.
The system displays the Route Policies page.
- Step 2** Click the underlined name of the route policy for which you want to see the route policy steps.

The system displays the Route Policy ‘<name of route policy>’ Steps page and displays all the steps associated with this route policy.

- Step 3** To delete a route policy step, do the following:
- Check the check box next to the name of the route policy step to delete.
 - Click **Remove**.
 - In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
- Step 4** To revert a route policy step to the settings it had at the time of the last commit, do the following:
- Check the check box next to the name of the route policy step whose settings you want to revert back to.
 - Click **Revert**.
 - In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Adding or Editing a Route Policy Step



Note When you edit a route policy, you can only edit the steps associated with it.

SUMMARY STEPS

- Choose **Configure > Route Policies**.
- Click the underlined name of the route policy for which you want to add or edit a route policy step.
- To add a route policy step, do the following:
- To edit a route policy step, do the following:
- To move a route policy step, check the check box next to it and click the up or down arrows.
- In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Route Policies**.
- The system displays the Route Policies page.
- Step 2** Click the underlined name of the route policy for which you want to add or edit a route policy step.
- The system displays the Route Policy Steps: <name of route policy> page and displays all the steps associated with this route policy.
- Step 3** To add a route policy step, do the following:
- Click **Add**. The system displays the Route Policy Step (New) page.
 - Enter information about the route policy step as described in [Route Policy Step Fields, on page 62](#).
 - Click **Add**.
- Step 4** To edit a route policy step, do the following:
- Click the underlined name of the route policy step. The system displays the Route Policy Step: *Edit* page.

- b) Make changes to the values for the route policy step as described in [Route Policy Step Fields, on page 62](#).
- c) Click **Update**.

Step 5 To move a route policy step, check the check box next to it and click the up or down arrows.

Step 6 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173



CHAPTER 10

Configuring Normalization Policies

- [Viewing a List of Normalization Policies, on page 69](#)
- [Adding a Normalization Policy, on page 83](#)
- [Working With URI Components for a Request URI, on page 84](#)
- [Working With URI Conversion Parameters for a Request URI, on page 85](#)
- [Working With URI Parameters for a Request URI, on page 85](#)
- [Working With SIP Headers, on page 87](#)
- [Working With URI Components for SIP Headers, on page 88](#)
- [Working With URI Conversion Parameters for SIP Headers, on page 89](#)
- [Working With URI Parameters for SIP Headers, on page 90](#)
- [Working With Header Parameters for SIP Headers, on page 91](#)

Viewing a List of Normalization Policies

SUMMARY STEPS

1. Choose **Configure** > **Normalization Policies**.
2. To delete a normalization policy, do the following:
3. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

Step 1 Choose **Configure** > **Normalization Policies**.

The system displays the Normalization Policies page, containing the fields described in the section Normalization Policy Fields.

Step 2 To delete a normalization policy, do the following:

- a) Check the check box next to the name of the normalization policy to delete.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Step 3 To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

- a) Check the check box next to the name of the normalization policy that has the changes to revert back to.

- b) Click **Revert**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

About Normalization Policies

Normalization policies modify SIP messages to account for incompatibilities between networks.

Normalization Policy Fields

The table lists the fields on the Normalization Policies page.

Table 15: Normalization Policy Parameters

| Parameter | Description |
|-----------|--|
| State | <p>Can be one of the following:</p> <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Name of this normalization policy. |

Request URI, URI Component Fields

The table lists the fields on the Normalization Policy '<name of normalization policy>' page when the Request URI and URI Component tabs are displayed.

Table 16: Request URI, URI Component Fields

| Parameter | Description |
|---------------|---|
| Category | <p>There are five boxes on this page, one for each of the following:</p> <ul style="list-style-type: none"> • User—Specifies the normalization policy to apply to the user URI component. • Phone—Specifies the normalization policy to apply to the phone URI component. • Host—Specifies the normalization policy to apply to the host URI component. • Host and Port—Specifies the normalization policy to apply to the host-port URI component. • URI—Specifies the normalization policy to apply to the full URI. <p>For each box, enter the match pattern and replace value.</p> |
| Match Pattern | Specifies the regular expression string in the URI component that is matched. If you enter all , the full header is replaced. |
| Replace Value | Specifies the regular expression string in the URI component that replaces the matched string. |

Related Topics

[Managing the System Configuration](#), on page 173

Request URI, URI Conversion Fields

The table lists the fields on the Normalization Policy ‘<name of normalization policy>’ page when the Request URI and URI Conversion tabs are displayed.

Table 17: Request URI, URI Conversion Fields

| Parameter | Description |
|--------------------------------------|--|
| SIP URI to TEL URI Conversion | |
| Conversion | Whether this conversion is enabled or disabled. The default is disabled. |
| TEL URI to SIP URI Conversion | |
| Conversion | Whether this conversion is enabled or disabled. The default is disabled. |

| Parameter | Description |
|-----------|--------------------------------|
| Host | Specifies the host of the URI. |
| Port | Specifies the port of the URI. |

Related Topics

[Managing the System Configuration](#), on page 173

Request URI, URI Parameter Fields

The table lists the fields on the Normalization Policy ‘<name of normalization policy>’ page when the Request URI and URI Parameter tabs are displayed.

Table 18: Request URI, URI Parameter Fields

| Parameter | Description |
|------------------------------|---|
| Add URI Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Specifies the URI parameter name to which the normalization rule applies. |
| Value | Specifies the value to be added to the URI parameter. |
| Remove URI Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Specifies the URI parameter name. |

| Parameter | Description |
|------------------------------|---|
| Update URI Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Specifies the header parameter name. |
| Match Pattern | Specifies the regular expression string in the URI parameter that is matched. If you enter all , the full header is replaced. |
| Replace Value | Specifies the regular expression string in the URI parameter that replaces the matched string. |

Related Topics

[Managing the System Configuration](#), on page 173

SIP Headers Fields

The table lists the fields on the Normalization Policy ‘<name of normalization policy>’ page when the SIP Header tabs are displayed.

Table 19: SIP Header Parameter Fields

| Parameter | Description |
|------------------------|---|
| Add SIP Headers | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |

| Parameter | Description |
|----------------------------------|---|
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Instances | The SIP header instances to be added. |
| Remove SIP Headers | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| Total Number of Header Instances | Total number of SIP header instances to be removed. |
| Update SIP Headers | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |

| Parameter | Description |
|------------------|---|
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied to all occurrences. |
| Match Pattern | Specifies the regular expression string in the header parameter that is matched. If you enter all , the full header is replaced. |
| Replace Value | Specifies the regular expression string in the header parameter that replaces the matched string. |

Related Topics

[Managing the System Configuration](#), on page 173

SIP Header, URI Component Fields

The table lists the fields on the Normalization Policy ‘<name of normalization policy>’ page when the SIP Header and URI Component tabs are displayed.

Table 20: SIP Header, URI Component Fields

| Parameter | Description |
|-----------|---|
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |

| Parameter | Description |
|--------------------|---|
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given URI component, apply this normalization step only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given URI component, apply this normalization step only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given URI component, apply this normalization step to all occurrences. |
| URI Component Type | Can be one of the following: <ul style="list-style-type: none"> • URI—Specifies the lookup policy to apply to the full URI. • User (default)—Specifies the lookup policy to apply to the user URI component. • Phone—Specifies the lookup policy to apply to the phone URI component. • Host—Specifies the lookup policy to apply to the host URI component. • Host-Port—Specifies the lookup policy to apply to the host-port URI component. |
| Match Pattern | Specifies the regular expression string in the URI component that is matched. If you enter all , the full header is replaced. |
| Replace Value | Specifies the regular expression string in the URI component that replaces the matched string. |

Related Topics

[Managing the System Configuration](#), on page 173

SIP Header, URI Conversion Fields

The table lists the fields on the Normalization Policy ‘<name of normalization policy>’ page when the SIP Header and URI Conversion tabs are displayed.

Table 21: SIP Header, URI Conversion Fields

| Parameter | Description |
|---------------------------------------|---|
| TEL URI to SIP URI Conversions | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given TEL URI, apply this normalization step only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given TEL URI, apply this normalization step only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given TEL URI, apply this normalization step to all occurrences. |
| Host | Specifies the host of the URI. |
| Port | Specifies the port of the URI. |
| SIP URI to TEL URI Conversions | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |

| Parameter | Description |
|------------------|--|
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a specific SIP URI, apply this normalization step only to the first occurrence. • last—Specifies that if there are multiple occurrences of a specific SIP URI, apply this normalization step only to the last occurrence. • all—Specifies that if there are multiple occurrences of a specific SIP URI, apply this normalization step to all occurrences. |

Related Topics

[Managing the System Configuration](#), on page 173

SIP Header, URI Parameter Fields

The table lists the fields on the Normalization Policy '<name of normalization policy>' page when the SIP Header and URI Parameter tabs are displayed.

Table 22: SIP Header, URI Parameter Fields

| Parameter | Description |
|---------------------------|---|
| Add URI Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |

| Parameter | Description |
|------------------------------|---|
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step to all occurrences. |
| Parameter Name | Specifies the URI parameter name to which the normalization rule applies. |
| Value | Specifies the value to be added. |
| Remove URI Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step to all occurrences. |

| Parameter | Description |
|------------------------------|---|
| Parameter Name | Specifies the URI parameter name. |
| Update URI Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given URI parameter, apply this normalization step to all occurrences. |
| Parameter Name | Specifies the header parameter name. |
| Match Pattern | Specifies the regular expression string in the URI parameter that is matched. If you enter all , the full header is replaced. |
| Replace Value | Specifies the regular expression string in the URI parameter that replaces the matched string. |

Related Topics

[Managing the System Configuration](#), on page 173

SIP Header, Header Parameter Fields

The table lists the fields on the Normalization Policy ‘<name of normalization policy>’ page when the SIP Header and Header Parameter tabs are displayed.

Table 23: SIP Header, Header Parameter Fields

| Parameter | Description |
|---------------------------------|---|
| Add Header Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied to all occurrences. |
| Parameter Name | Name of this add URI parameter. |
| Value | Value of the add URI parameter. |
| Remove Header Parameters | |

| Parameter | Description |
|---------------------------------|---|
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied to all occurrences. |
| Parameter Name | Name of this remove URI parameter. |
| Update Header Parameters | |
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |

| Parameter | Description |
|------------------|---|
| SIP Header Name | Specifies the SIP message header for which the normalization step is applicable. Examples include: From, To, Record-Route, Diversion, Request-URI, and P-Asserted-Identity. |
| SIP Header Index | Can be one of the following: <ul style="list-style-type: none"> • first—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the first occurrence. • last—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied only to the last occurrence. • all—Specifies that if there are multiple occurrences of a given header parameter, this normalization step is applied to all occurrences. |
| Parameter Name | Name of this update URI parameter. |
| Match Pattern | Specifies the regular expression string in the URI component that is matched. If you enter all , the full header is replaced. |
| Replace Value | Specifies the regular expression string in the URI component that replaces the matched string. |

Related Topics

[Managing the System Configuration](#), on page 173

Adding a Normalization Policy

SUMMARY STEPS

1. Choose **Configure > Normalization Policies**.
2. Click **Add**.
3. Enter a name for this normalization policy.
4. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure > Normalization Policies**.

The system displays the Normalization Policies page.

Step 2 Click **Add**.

The system displays the Normalization Policies page.

Step 3 Enter a name for this normalization policy.

Click **Add**.

The system displays the Normalization Policies page, with the new normalization policy listed.

Note Ensure not to include double quotes while configuring Normalization Policies from GUI.

Step 4 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Working With URI Components for a Request URI

SUMMARY STEPS

1. Choose **Configure > Normalization Policies**.
2. Click the underlined name of the normalization policy to work with.
3. To add or edit a URI component, do the following:
4. To delete a URI component, do the following:
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure > Normalization Policies**.

The system displays the Normalization Policies page.

Step 2 Click the underlined name of the normalization policy to work with.

The system displays the Normalization Policy '<name of normalization policy>' page and the URI Component tab is highlighted.

Step 3 To add or edit a URI component, do the following:

- a) Check the check box of the component to which you want to add or edit values.
- b) Enter or change values. See [Request URI, URI Component Fields, on page 70](#).
- c) Click **Update**.

Step 4 To delete a URI component, do the following:

- a) Uncheck the check box of the component to delete.
- b) Click **Update**.

Step 5 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Working With URI Conversion Parameters for a Request URI

Follow this procedure to configure a normalization policy step that converts a destination TEL URI to a SIP URI with the given host-port value.

SUMMARY STEPS

1. Choose **Configure > Normalization Policies**.
2. Click the underlined name of the normalization policy to work with.
3. Click the URI Conversion tab.
4. Enter or update values. See [Request URI, URI Conversion Fields, on page 71](#).
5. Click **Update**.
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Normalization Policies**.
The system displays the Normalization Policies page.
- Step 2** Click the underlined name of the normalization policy to work with.
The system displays the Normalization Policy ‘<name of normalization policy>’ page.
- Step 3** Click the URI Conversion tab.
- Step 4** Enter or update values. See [Request URI, URI Conversion Fields, on page 71](#).
- Step 5** Click **Update**.
- Step 6** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173

Working With URI Parameters for a Request URI

SUMMARY STEPS

1. Choose **Configure > Normalization Policies**.
2. Click the underlined name of the normalization policy to work with.
3. Click the URI Parameter tab.
4. To add a URI parameter to the Request URI, do the following:
5. To remove a parameter from the URI, do the following:
6. To update a parameter in the URI, do the following:

7. To remove any parameters that you added in **Step 4** to Step 6, check the check box next to the parameter and click **Remove**.
8. To revert to the previous setting for any parameters that you added in **Step 4** to **Step 6**, check the check box next to the parameter and click **Revert**.
9. To edit the add or update parameters that you added in **Step 4** or **Step 6**, click the name of the parameter and make changes.
10. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure > Normalization Policies**.
The system displays the Normalization Policies page.
- Step 2** Click the underlined name of the normalization policy to work with.
The system displays the Normalization Policy '**<name of normalization policy>**' page.
- Step 3** Click the URI Parameter tab.
- Step 4** To add a URI parameter to the Request URI, do the following:
- a) Under the Add URI Parameters heading, click **New**.
 - b) Enter the name of the parameter and a value.
 - c) Click **Add**.
- Step 5** To remove a parameter from the URI, do the following:
- a) Under the Remove URI Parameters heading, click **New**.
 - b) Enter the name of the parameter to remove.
 - c) Click **Add**.
- Step 6** To update a parameter in the URI, do the following:
- a) Under the Update URI Parameters heading, click **New**.
 - b) Enter the name of the parameter to update and the pattern to match. Optionally, you can enter a value to replace the pattern.
 - c) Click **Add**.
- Step 7** To remove any parameters that you added in **Step 4** to Step 6, check the check box next to the parameter and click **Remove**.
- Step 8** To revert to the previous setting for any parameters that you added in **Step 4** to **Step 6**, check the check box next to the parameter and click **Revert**.
- Step 9** To edit the add or update parameters that you added in **Step 4** or **Step 6**, click the name of the parameter and make changes.
- Step 10** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Working With SIP Headers

SUMMARY STEPS

1. Choose **Configure > Normalization Policies**.
2. Click the underlined name of the normalization policy to which you want to add a SIP header.
3. Click the SIP Header tab.
4. To add a SIP header, do the following:
5. To remove a SIP header, do the following:
6. To update a SIP header, do the following:
7. To remove any SIP headers that you added in **Step 4** to **Step 6**, check the check box next to the parameter and click **Remove**.
8. To revert to the previous setting for any SIP headers that you added in **Step 4** to **Step 6**, check the check box next to the SIP header and click **Revert**.
9. To edit the add or update parameters that you added in **Step 4** or **Step 6**, click the name of the SIP header and make changes.
10. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure > Normalization Policies**.

The system displays the Normalization Policies page.

Step 2 Click the underlined name of the normalization policy to which you want to add a SIP header.

The system displays the Normalization Policy '<name of normalization policy>' page.

Step 3 Click the SIP Header tab.

The system displays the Normalization Policy '<name of normalization policy>' page with the SIP Header tabs displayed.

Step 4 To add a SIP header, do the following:

- a) Under the Add SIP Headers heading, click **New**.
- b) Enter the name of the parameter.
- c) Click **Add**.
- d) Enter a SIP header index and value.
- e) Click **Add**.
- f) Click **Cancel** to go back to the Normalization Policy: <name of normalization policy> page with the SIP Header tabs displayed.

Step 5 To remove a SIP header, do the following:

- a) Under the Remove SIP Headers heading, click **New**.
- b) Enter the name of the SIP header to remove. Enter the number of header instances to be removed from the top and the number to be removed from the bottom.
- c) Click **Add**.

Step 6 To update a SIP header, do the following:

- a) Under the Update SIP Headers heading, click **New**.
- b) Enter the name of the SIP header to update and the pattern to match. You can optionally enter a SIP header index and a value to replace the pattern with.
- c) Click **Add**.

- Step 7** To remove any SIP headers that you added in **Step 4** to **Step 6**, check the check box next to the parameter and click **Remove**.
- Step 8** To revert to the previous setting for any SIP headers that you added in **Step 4** to **Step 6**, check the check box next to the SIP header and click **Revert**.
- Step 9** To edit the add or update parameters that you added in **Step 4** or **Step 6**, click the name of the SIP header and make changes.
- Step 10** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Working With URI Components for SIP Headers

Follow this procedure to configure a normalization policy step that updates a URI component field within a header of the source message.

SUMMARY STEPS

1. Choose **Configure > Normalization Policies**.
2. Click the underlined name of the normalization policy to work with.
3. Click the SIP Header tab.
4. Click the URI Component tab.
5. To add a URI component to a SIP header, do the following:
6. To edit a URI component for a SIP header, do the following:
7. To remove a URI component for a SIP header, check the check box next to the URI component and click **Remove**.
8. To revert to the previous setting for a URI component for a SIP header, check the check box next to the URI component and click **Revert**.
9. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Normalization Policies**.
The system displays the Normalization Policies page.
- Step 2** Click the underlined name of the normalization policy to work with.
The system displays the Normalization Policy ‘<name of normalization policy>’ page.
- Step 3** Click the SIP Header tab.
- Step 4** Click the URI Component tab.

- Step 5** To add a URI component to a SIP header, do the following:
- Click **New**.
 - Enter values. See [SIP Header, URI Component Fields, on page 75](#).
 - Click **Add**.
- Step 6** To edit a URI component for a SIP header, do the following:
- Click the underlined name of the SIP header.
 - Update the match pattern or replace values. See [SIP Header, URI Component Fields, on page 75](#).
 - Click **Update**.
- Step 7** To remove a URI component for a SIP header, check the check box next to the URI component and click **Remove**.
- Step 8** To revert to the previous setting for a URI component for a SIP header, check the check box next to the URI component and click **Revert**.
- Step 9** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Working With URI Conversion Parameters for SIP Headers

SUMMARY STEPS

- Choose **Configure > Normalization Policies**.
- Click the underlined name of the normalization policy to work with.
- Click the SIP Header tab.
- Click the URI Conversion tab.
- To add a new conversion parameter, do the following:
- To edit a TEL URI to SIP URI conversion parameter, do the following:
- To remove a URI conversion parameter, check the check box next to the URI conversion parameter and click **Remove**.
- To revert to the previous setting for a URI conversion parameter, check the check box next to the URI conversion parameter and click **Revert**.
- In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Normalization Policies**.
The system displays the Normalization Policies page.
- Step 2** Click the underlined name of the normalization policy to work with.
The system displays the Normalization Policy ‘<name of normalization policy>’ page.
- Step 3** Click the SIP Header tab.
- Step 4** Click the URI Conversion tab.
- Step 5** To add a new conversion parameter, do the following:

- a) Click **New** under either the TEL URI to SIP URI Conversions header or the SIP URI to TEL URI Conversions header.
- b) Enter values. See [SIP Header, URI Conversion Fields, on page 76](#) the section SIP Header, URI Conversion Fields.
- c) Click **Add**.

Step 6 To edit a TEL URI to SIP URI conversion parameter, do the following:

- a) Click the underlined name of the SIP header.
- b) Update values. See [SIP Header, URI Conversion Fields, on page 76](#).
- c) Click **Update**.

Step 7 To remove a URI conversion parameter, check the check box next to the URI conversion parameter and click **Remove**.

Step 8 To revert to the previous setting for a URI conversion parameter, check the check box next to the URI conversion parameter and click **Revert**.

Step 9 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Working With URI Parameters for SIP Headers

SUMMARY STEPS

1. Choose **Configure > Normalization Policies**.
2. Click the underlined name of the normalization policy to work with.
3. Click the SIP Header tab.
4. Click the URI Parameter tab.
5. To add a URI parameter to the SIP header do the following:
6. To remove a URI parameter from the SIP header, do the following:
7. To update a URI parameter in the SIP header, do the following:
8. To remove any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Remove**.
9. To revert to the previous setting for any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Revert**.
10. To edit the add or update parameters that you added in **Step 5** or **Step 7**, click the name of the parameter and make changes.
11. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure > Normalization Policies**.

The system displays the Normalization Policies page.

Step 2 Click the underlined name of the normalization policy to work with.

The system displays the Normalization Policy '<name of normalization policy>' page.

Step 3 Click the SIP Header tab.

- Step 4** Click the URI Parameter tab.
- Step 5** To add a URI parameter to the SIP header do the following:
- Under the Add URI Parameters heading, click **New**.
 - Enter values. See [SIP Header, URI Parameter Fields, on page 78](#) the section SIP Header, URI Parameter Fields.
 - Click **Add**.
- Step 6** To remove a URI parameter from the SIP header, do the following:
- Under the Remove URI Parameters heading, click **New**.
 - Enter values. See [SIP Header, URI Parameter Fields, on page 78](#).
 - Click **Add**.
- Step 7** To update a URI parameter in the SIP header, do the following:
- Under the Update URI Parameters heading, click **New**.
 - Enter values. See [SIP Header, URI Parameter Fields, on page 78](#).
 - Click **Add**.
- Step 8** To remove any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Remove**.
- Step 9** To revert to the previous setting for any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Revert**.
- Step 10** To edit the add or update parameters that you added in **Step 5** or **Step 7**, click the name of the parameter and make changes.
- Step 11** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Working With Header Parameters for SIP Headers

SUMMARY STEPS

- Choose **Configure > Normalization Policies**.
- Click the underlined name of the normalization policy to work with.
- Click the SIP Header tab.
- Click the Header Parameter tab.
- To add a header parameter to the SIP header do the following:
- To remove a header parameter from the SIP header, do the following:
- To update a header parameter in the SIP header, do the following:
- To remove any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Remove**.
- To revert to the previous setting for any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Revert**.
- To edit the add or update parameters that you added in **Step 5** or **Step 7**, click the name of the parameter and make changes.
- In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

- Step 1** Choose **Configure** > **Normalization Policies**.
The system displays the Normalization Policies page.
- Step 2** Click the underlined name of the normalization policy to work with.
The system displays the Normalization Policy '<name of normalization policy>' page.
- Step 3** Click the SIP Header tab.
- Step 4** Click the Header Parameter tab.
- Step 5** To add a header parameter to the SIP header do the following:
- Under the Add Header Parameters heading, click **New**.
 - Enter values. See [SIP Header, Header Parameter Fields, on page 80](#).
 - Click **Add**.
- Step 6** To remove a header parameter from the SIP header, do the following:
- Under the Remove Header Parameters heading, click **New**.
 - Enter values. See [SIP Header, Header Parameter Fields, on page 80](#).
 - Click **Add**.
- Step 7** To update a header parameter in the SIP header, do the following:
- Under the Update Header Parameters heading, click **New**.
 - Enter values. See [SIP Header, Header Parameter Fields, on page 80](#).
 - Click **Add**.
- Step 8** To remove any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Remove**.
- Step 9** To revert to the previous setting for any parameters that you added in **Step 5** to **Step 7**, check the check box next to the parameter and click **Revert**.
- Step 10** To edit the add or update parameters that you added in **Step 5** or **Step 7**, click the name of the parameter and make changes.
- Step 11** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173



CHAPTER 11

Configuring Time Policies

- [Viewing a List of Time Policies](#), on page 93
- [Adding a Time Policy](#), on page 94
- [Viewing a List of Time Policy Steps](#), on page 95
- [Adding or Editing a Time Policy Step](#), on page 95

Viewing a List of Time Policies

SUMMARY STEPS

1. Choose **Configure** > **Time Policies**.
2. To delete a time policy, do the following:
3. To revert any changes you have made back to the state they were in at the time of the last commit, do the following:

DETAILED STEPS

-
- Step 1** Choose **Configure** > **Time Policies**.
- The system displays the Time Policies page showing the time policies with the fields in [Time Policy Fields](#), on page 94.
- Step 2** To delete a time policy, do the following:
- a) Check the check box next to the name of the time policy to delete.
 - b) Click **Remove**.
 - c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
- Step 3** To revert any changes you have made back to the state they were in at the time of the last commit, do the following:
- a) Check the check box next to the name of the time policy that has the changes to revert back to.
 - b) Click **Revert**.
 - c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Related Topics

[Managing the System Configuration](#), on page 173

About Time Policies

Time policies are time-based routing configurations that a route group will use if implementing time-based routing.

Time Policy Fields

The table lists the fields on the Time Policies page.

Table 24: Time Policy Parameters

| Parameter | Description |
|-----------|---|
| State | Can be one of the following: <ul style="list-style-type: none"> • New—New record. Will be added to the active configuration when it is committed. • Modified—Modified record. Will become the active configuration when it is committed. • Deleted—Deleted record. Will be removed from the active configuration when it is committed. • Active—Active record and active configuration. |
| Name | Name of this time policy. |

Adding a Time Policy

SUMMARY STEPS

1. Choose **Configure > Time Policies**.
2. Click **Add**.
3. Enter a name for this time policy.
4. Add steps to the time policy. See [Adding or Editing a Time Policy Step, on page 95](#).
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

-
- Step 1** Choose **Configure > Time Policies**.
The system displays the Time Policies page.
- Step 2** Click **Add**.
The system displays the Time Policy (New) page.
- Step 3** Enter a name for this time policy.

Click **Add**.

The system displays the Time Policy '<name of time policy>' Step (New) page.

Step 4 Add steps to the time policy. See [Adding or Editing a Time Policy Step, on page 95](#).

Step 5 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Viewing a List of Time Policy Steps

SUMMARY STEPS

1. Choose **Configure > Time Policies**.
2. Click the underlined name of a time policy.

DETAILED STEPS

Step 1 Choose **Configure > Time Policies**.

The system displays the Time Policies page.

Step 2 Click the underlined name of a time policy.

The system displays the Time Policy '<name of time policy>' Step page.

Adding or Editing a Time Policy Step

SUMMARY STEPS

1. Choose **Configure > Time Policies**.
2. Click the underlined name of a time policy.
3. To add a time policy step, do the following:
4. To edit a time policy step, do the following:
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure > Time Policies**.

The system displays the Time Policies page.

Step 2 Click the underlined name of a time policy.

The system displays the Time Policy '<name of time policy>' Steps page.

- Step 3** To add a time policy step, do the following:
- Click **Add**. The system displays the Time Policy '<name of time policy>' Step (New) page.
 - Enter values in the fields. See [Time Policy Steps, on page 96](#) the section Time Policy Steps.
 - Click **Update**.
- Step 4** To edit a time policy step, do the following:
- Click the underlined name of a time policy step. The system displays the Time Policy '<name of time policy>' Step page.
 - Update values in the fields.
 - Click **Update**.
- Step 5** In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Time Policy Steps

Table 25: Time Policy Steps

| Parameter | Description |
|------------------------------|---|
| Active Dates | |
| Start Date & Time | Start date and time of this time policy. Enter the date, hour, minute, and either AM or PM. |
| End Date & Time | End date and time of this time policy. If you check this check box and click Update , the system prompts you to enter a date. |
| Schedule Restrictions | |
| Weekdays/Dates | Defines any weekday or date restrictions that your time policy may have. If you check this check box and click Update , the system prompts you to choose either Days of the Week or Days of the Month. <ul style="list-style-type: none"> If you check Days of the Week, the system prompts you to check which days of the week this policy covers. If you check Days of the Month, the system prompts you to check which days of the month this policy covers. |

| Parameter | Description |
|-------------|---|
| Months | <p>Defines any monthly restrictions that your time policy may have.</p> <p>If you check this check box and click Update, the system prompts you to check which months this policy covers.</p> |
| Time of Day | <p>Defines any time of day restrictions that your time policy may have.</p> <p>If you check this check box and click Update, the system prompts you to enter a time. After you enter a time, click Add. You can enter additional times.</p> |



CHAPTER 12

Configuring Routing Triggers

- [Viewing a List of Routing Triggers, on page 99](#)
- [Adding or Editing a Routing Trigger, on page 100](#)

Viewing a List of Routing Triggers

Routing triggers correlate trigger conditions with routing policies (which are also known as lookup policies). A single policy is chosen based on which corresponding condition is matched. The conditions are evaluated in ascending order based on sequence number.

A routing trigger is a set of conditions that can be used to dictate routing logic. It is automatically executed in response to a certain event (or condition case). Conditions can have multiple cases.

SUMMARY STEPS

1. Choose **Configure > Routing Triggers**.
2. To delete a routing trigger, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Routing Triggers**.

The system displays the Routing Triggers page and displays all routing triggers.

Step 2 To delete a routing trigger, do the following:

- a) Check the check box next to the name of the routing trigger to delete.
 - b) Click **Remove**.
 - c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.
-

Adding or Editing a Routing Trigger

Before you begin

You must have at least one trigger in your system. See [Configuring Triggers, on page 19](#).

SUMMARY STEPS

1. Choose Configure > **Routing** Triggers.
2. To add a routing trigger, do the following:
3. To edit an existing routing trigger, do the following:
4. To move an existing routing trigger, do the following:
5. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose Configure > **Routing** Triggers.

The system displays the Routing Triggers page.

Step 2 To add a routing trigger, do the following:

- a) Click **Add**.
- b) The system displays the Routing Trigger (New) page.
- c) Select a routing policy from the drop-down box.
- d) Select a trigger condition from the drop-down box.
- e) Click **Add**.

The system displays the Routing Triggers page with the new routing trigger displayed.

Step 3 To edit an existing routing trigger, do the following:

- a) Check the check box next to the name of the routing trigger to edit.
- b) Click **Edit**.
- c) Choose a different routing policy or trigger condition. You can change one or both.
- d) Click **Update**.

Step 4 To move an existing routing trigger, do the following:

- a) Check the check box next to the name of the routing trigger to move.
- b) Click the up or down arrows.

Step 5 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Configuring Routing Triggers, on page 99](#)

[Managing the System Configuration, on page 173](#)



CHAPTER 13

Configuring Normalization Triggers

- [Viewing a List of Pre-Normalization Triggers, on page 101](#)
- [Viewing a List of Post-Normalization Triggers, on page 102](#)
- [Adding and Editing a Pre-Normalization Trigger, on page 102](#)
- [Adding and Editing a Post-Normalization Trigger, on page 103](#)

Viewing a List of Pre-Normalization Triggers

SUMMARY STEPS

1. Choose **Configure** > **Normalization Triggers** > **Pre-Normalization**.
2. To delete a pre-normalization trigger, do the following:

DETAILED STEPS

Step 1 Choose **Configure** > **Normalization Triggers** > **Pre-Normalization**.

The system displays the Pre-Normalization Triggers page and displays all pre-normalization triggers.

Step 2 To delete a pre-normalization trigger, do the following:

- a) Check the check box next to the name of the pre-normalization trigger to delete.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

About Normalization Triggers

Normalization triggers correlate trigger conditions with normalization policies. There are two types of normalization triggers:

- pre-normalization, which occur before routing
- post-normalization, which occur after routing

A special policy bypasses normalization on mid-dialog messages.

You can add, update, or delete normalization triggers from the Pre-Normalization Triggers and Post-Normalization Triggers pages.

Related Topics

[Managing the System Configuration](#), on page 173

Viewing a List of Post-Normalization Triggers

For information on normalization triggers, see [About Normalization Triggers](#), on page 101.

SUMMARY STEPS

1. Choose **Configure** > **Normalization Triggers** > **Post-Normalization**.
2. To delete a post-normalization trigger, do the following:

DETAILED STEPS

Step 1 Choose **Configure** > **Normalization Triggers** > **Post-Normalization**.

The system displays the Post-Normalization Triggers page and displays all post-normalization triggers.

Step 2 To delete a post-normalization trigger, do the following:

- a) Check the check box next to the name of the post-normalization trigger to delete.
- b) Click **Remove**.
- c) In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Managing the System Configuration](#), on page 173

Adding and Editing a Pre-Normalization Trigger

SUMMARY STEPS

1. Choose **Configure** > **Normalization Triggers** > **Pre-Normalization**.
2. To add a pre-normalization trigger, do the following:
3. To add, edit, or delete rules for a pre-normalization trigger, follow the procedure in [Viewing, Adding, Moving, and Deleting Rules for a Trigger](#), on page 27.
4. To edit a pre-normalization trigger, do the following:
5. If you have multiple pre-normalization triggers, you can reorder them by doing the following:
6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure** > **Normalization Triggers** > **Pre-Normalization**.

The system displays the Pre-Normalization Triggers page.

Step 2 To add a pre-normalization trigger, do the following:

- a) Click **Add**. The system displays the Pre-Normalization Trigger (New) page.
- b) Choose a normalization policy from the drop-down menu.
- c) Choose a trigger condition from the drop-down menu.
- d) Click **Add**.

The system displays the Pre-Normalization Triggers page and displays all of the triggers.

Step 3 To add, edit, or delete rules for a pre-normalization trigger, follow the procedure in [Viewing, Adding, Moving, and Deleting Rules for a Trigger, on page 27](#).

Step 4 To edit a pre-normalization trigger, do the following:

- a) Check the check box of the pre-normalization trigger to edit.
- b) Click **Edit**. The system displays the Pre-Normalization Trigger page.
- c) Choose a normalization policy from the drop-down menu.
- d) Choose a trigger condition from the drop-down menu.
- e) Click **Update**. The system displays the Pre-Normalization Triggers page and displays all of the triggers.

Step 5 If you have multiple pre-normalization triggers, you can reorder them by doing the following:

Tip Once one pre-normalization trigger is matched, all other triggers are ignored. To optimize the system, we recommend that you put the pre-normalization trigger most likely to match at the top of the list.

- a) Select the pre-normalization trigger.
- b) Click the up or down arrows.
- c) Click **Update**.

Step 6 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Configuring Triggers](#), on page 19

[Managing the System Configuration](#), on page 173

Adding and Editing a Post-Normalization Trigger

SUMMARY STEPS

1. Choose **Configure** > **Normalization Triggers** > **Post-Normalization**.
2. To add a post-normalization trigger, do the following:
3. To add, edit, or delete rules for a post-normalization trigger, follow the procedure in [Viewing, Adding, Moving, and Deleting Rules for a Trigger, on page 27](#).
4. To edit a post-normalization trigger, do the following:
5. If you have multiple post-normalization triggers, you can reorder them by doing the following:

6. In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

DETAILED STEPS

Step 1 Choose **Configure > Normalization Triggers > Post-Normalization**.

The system displays the Post-Normalization Triggers page.

Step 2 To add a post-normalization trigger, do the following:

- a) Click **Add**. The system displays the Post-Normalization Trigger (New) page.
- b) Choose a normalization policy from the drop-down menu.
- c) Choose a trigger condition from the drop-down menu.
- d) Click **Add**.

The system displays the Post-Normalization Triggers page and displays all of the triggers.

Step 3 To add, edit, or delete rules for a post-normalization trigger, follow the procedure in [Viewing, Adding, Moving, and Deleting Rules for a Trigger, on page 27](#).

Step 4 To edit a post-normalization trigger, do the following:

- a) Check the check box of the post-normalization trigger to edit.
- b) Click **Edit**. The system displays the Post-Normalization Trigger page.
- c) Choose a normalization policy from the drop-down menu.
- d) Choose a trigger condition from the drop-down menu.
- e) Click **Update**. The system displays the Post-Normalization Triggers page and displays all of the triggers.

Step 5 If you have multiple post-normalization triggers, you can reorder them by doing the following:

Tip Once one post-normalization trigger is matched, all other triggers are ignored. To optimize the system, we recommend that you put the post-normalization trigger most likely to match at the top of the list.

- a) Select the post-normalization trigger.
- b) Click the up or down arrows.
- c) Click **Update**.

Step 6 In the Cisco Unified SIP Proxy header, click **Commit Candidate Configuration** to commit this change.

Related Topics

[Configuring Triggers](#), on page 19

[Managing the System Configuration](#), on page 173



CHAPTER 14

Configuring Performance Control

- [Configuring Performance Control, on page 105](#)

Configuring Performance Control

Use this page to enable or disable Lite Mode and to set the maximum number of calls per second that the system can process.



Restriction

- If you enable Lite Mode, the system deletes the record route configurations and you cannot access the SIP Record-Route tab. For information about the SIP Record-Route tab, see [Editing the SIP Record-Route for a Network, on page 17](#).
- Because call admission control relies on record-route, call admission control is disabled whenever Lite Mode is enabled.

SUMMARY STEPS

1. Choose **Configure > Performance Control**.
2. Select if you want to enable or disable Lite Mode:
3. (Optional) Enter the maximum limit for the calls per second on the system:

DETAILED STEPS

Step 1 Choose **Configure > Performance Control**.

The system displays the Performance Control page.

Step 2 Select if you want to enable or disable Lite Mode:

- Select **enable (<license limit> CPS)** to enable Lite Mode, which allows the system to process the number of calls up to the limit which is based on the license type. If you choose this option, the system asks you to confirm that you want to enter Lite Mode, which will disable record-routing. Click **OK**.

- Select **disable** (<license limit> **CPS**) to disable Lite Mode, which limits the system to only processing the number of calls up to the limit. If you choose this option, the system asks you to confirm that you want to disable Lite Mode, which will reset performance to licensed limits. Click **OK**.

Step 3 (Optional) Enter the maximum limit for the calls per second on the system:

- If you selected **enable** (<license limit> **CPS**) to enable Lite Mode, the value must be the value of the license limit or less. Click **Set Limit**.
 - If you selected **disable** (<license limit> **CPS**) to disable Lite Mode, the value must be the value of the licensed limit or less. Click **Set Limit**.
-



CHAPTER 15

Configuring Call Admission Control

- [Configuring Call Admission Control, on page 107](#)

Configuring Call Admission Control

The call admission control feature allows you to count and limit the number of calls for a certain location. This can only be performed for server group elements.

When call admission control is enabled, the system monitors the start and stop time for each call. You can also set the session timeout which tells the system how long to wait before a call is considered dead.

For call admission control to work correctly, record route needs to be enabled on Cisco Unified SIP Proxy. If record route is not enabled, call admission control will not work reliably.

SUMMARY STEPS

1. Choose **Configure > Call Admission Control**.
2. Select if you want to enable or disable Call Admission Control.
3. Enter the Call Admission Control session timeout in minutes.
4. Click **Update**.

DETAILED STEPS

Step 1 Choose **Configure > Call Admission Control**.

The system displays the Call Admission Control page.

Step 2 Select if you want to enable or disable Call Admission Control.

Step 3 Enter the Call Admission Control session timeout in minutes.

Note If call admission control is enabled and you change the configuration value, the system only uses the updated value for new calls. Any existing calls will continue to use the session timeout value that was configured when those calls were originally set up. Changing the session timeout has no effect on the timeout for existing, active calls.

Step 4 Click **Update**.



CHAPTER 16

Configuring Users

- [Viewing a List of Users, on page 109](#)
- [Adding a New User, on page 111](#)
- [Displaying or Changing a User Profile, on page 112](#)
- [Displaying or Changing Group Subscriptions, on page 112](#)
- [Finding a User, on page 113](#)
- [Changing Your Password, on page 114](#)

Viewing a List of Users



Important A user can only be subscribed as a member of either any of the default group, or one or more newly created groups.

You can delete a user, who is subscribed as member of non-default group, only on unsubscribing from the associated non-default groups.

SUMMARY STEPS

1. Choose **Configure > Users**.
2. To delete a user from the Cisco Unified SIP Proxy system, do the following:
3. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
4. To sort users, click any of the headers.
5. To delete a user from the Cisco Unified SIP Proxy system, do the following:

DETAILED STEPS

Step 1 Choose **Configure > Users**.

The system displays the Configure Users page, containing the following fields:

- User ID—By default, the system displays users in alphabetical order by user ID.
- Display Name

- Primary Extension

Step 2 To delete a user from the Cisco Unified SIP Proxy system, do the following:

- Check the check box next to the user ID to delete.
- Click **Delete**.
- Click **OK** to confirm the deletion.

Step 3 To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.

Step 4 To sort users, click any of the headers.

Step 5 To delete a user from the Cisco Unified SIP Proxy system, do the following:

- Check the check box next to the user ID to delete.
- Click **Delete**.
- Click **OK** to confirm the deletion.

User Profile Fields

The table lists the fields on the User Profile page.

Table 26: User Profile Parameters

| Parameter | Description |
|----------------------|---|
| User ID | Alphanumeric user identifier. |
| First Name | First name of a user. Callers use these names to access the extension using the dial-by-name feature. These fields cannot contain special characters, spaces, or numbers. |
| Last Name | Last name of a user. Callers use these names to access the extension using the dial-by-name feature. These fields cannot contain special characters, spaces, or numbers. |
| Nick Name | Optional nickname of the user. |
| Display Name | User's name displayed within Cisco Unified SIP Proxy application. |
| Primary E.164 Number | User's primary telephone number, including area code. |
| Fax Number | Fax number for this user. |
| Language | The languages available depends on the version of Cisco Unified SIP Proxy that you have installed. |

| Parameter | Description |
|------------------|---|
| Password options | For the password used by the user to access the GUI, select one of the following: <ul style="list-style-type: none"> • Generate a Random Password—To have the system generate a random password. • Blank Password—To leave the password blank. • Password Specified Below—To specify a password for this user. |
| Password | Consists of letters and numbers and is at least 3 characters but not more than 32 characters long. |
| PIN options | Note Although there is space to set a PIN, the Cisco Unified SIP Proxy system does not use PINs. If you set values here, they will not be used. |
| PIN | Not used. |

Adding a New User

Use this procedure to add a new user to the system.



Note Ensure not to use reserved keywords or any name that may conflict with the regular users or process names in Linux (root,bin,daemon,adm,lp,sync,shutdown,halt,mail,operator,games, and so on.) while creating the users.

SUMMARY STEPS

1. Choose **Configure > Users**.
2. Click **Add**. The Add a New User window appears.
3. Enter information into the following fields. See [User Profile Fields, on page 110](#).
4. Click **Add**.

DETAILED STEPS

-
- Step 1** Choose **Configure > Users**.
The system displays the Configure Users page.
- Step 2** Click **Add**. The Add a New User window appears.
- Step 3** Enter information into the following fields. See [User Profile Fields, on page 110](#).
- Step 4** Click **Add**.

Note If you selected a random password or PIN, a message appears with the new password or PIN. Write these values in a secure place to give to the user. They are The value is also displayed on the user profile page (see [Displaying or Changing a User Profile, on page 112](#)).

Displaying or Changing a User Profile

The system displays the User Profile page, containing the fields in the section User Profile Fields.

SUMMARY STEPS

1. Select **Configure > Users**.
2. Click the underlined user ID of the person whose profile you want to see.

DETAILED STEPS

Step 1 Select **Configure > Users**.

The system displays the Configure Users page.

Step 2 Click the underlined user ID of the person whose profile you want to see.

Note If you do not see the user you are looking for, click **Find**. (See [Finding a User, on page 113](#).)

Related Topics

[Finding a User, on page 113](#)

Displaying or Changing Group Subscriptions

Use this procedure to modify the groups to which a user is assigned.

SUMMARY STEPS

1. Choose **Configure > Users**.
2. Click the underlined name of the user whose group subscription you want to view or modify.
3. Click the **Groups** tab. The following fields are displayed:
4. To subscribe the user as the owner of another group, click **Subscribe as owner**. To subscribe the user as a member of another group, click **Subscribe as member**.
5. Enter the group ID, description, or extension number, and click **Find**.
6. Check the check box next to the group for this user to join and click **Select Rows**.
7. (Optional) To unsubscribe the user from a group, check the check box next to the group name and click **Unsubscribe**.

DETAILED STEPS

- Step 1** Choose **Configure > Users**.
The system displays the Configure Users page.
- Step 2** Click the underlined name of the user whose group subscription you want to view or modify.
The system displays the User Profile page.
- Step 3** Click the **Groups** tab. The following fields are displayed:
- Group ID
 - Rights—whether the user is a member or owner of the group.
 - Description
 - Primary extension—primary extension assigned to the group.
- Step 4** To subscribe the user as the owner of another group, click **Subscribe as owner**. To subscribe the user as a member of another group, click **Subscribe as member**.
The system displays the Find page.
- Step 5** Enter the group ID, description, or extension number, and click **Find**.
- Step 6** Check the check box next to the group for this user to join and click **Select Rows**.
- Step 7** (Optional) To unsubscribe the user from a group, check the check box next to the group name and click **Unsubscribe**.

Related Topics

[Configuring Groups](#), on page 117

Finding a User

SUMMARY STEPS

1. Choose **Configure > Users**.
2. Click **Find**.
3. Enter the search criteria in one or more fields and click **Find**.

DETAILED STEPS

- Step 1** Choose **Configure > Users**.
The system displays the Configure Users window.
- Step 2** Click **Find**.
The system displays the following fields:
- User ID

- Name
- Extension

Step 3 Enter the search criteria in one or more fields and click **Find**.

The system displays the results of your search.

Changing Your Password



Restriction

- Passwords should be at least eight to 64 alphanumeric characters in length. However, there is no limit on the maximum length of the password. The administrator has the privileges to provision a user with password length that is higher than the minimum length. For example, a user can have password of length 65 alphanumeric characters.
- Use a mixture of uppercase and lowercase letters, symbols, and numbers.
- Spaces are not allowed.

SUMMARY STEPS

1. Select **Configure > Users**.
2. Click your name in the list of users.
3. Ensure that **Password specified below** is selected in the Password options field.
4. Enter your new password.
5. Enter your new password again for verification.
6. Click **Apply**.

DETAILED STEPS

Step 1 Select **Configure > Users**.
The system displays the Configure Users page.

Step 2 Click your name in the list of users.

Step 3 Ensure that **Password specified below** is selected in the Password options field.

Step 4 Enter your new password.

Step 5 Enter your new password again for verification.

Step 6 Click **Apply**.



CHAPTER 17

Setting User Defaults

- [User Defaults](#), on page 115
- [Configuring Password Options](#), on page 115
- [Configuring Account Lockout Policy](#), on page 116

User Defaults

When you create a user, the defaults that you set in the Configure User window take effect. Use these procedures to specify the default global password and PIN policy settings for all users. This default set of parameters is applied when a new user is created.

Perform the following tasks from the Configure User Defaults window:



Note Even after you have set defaults in this window, you can change the password policy for an individual user. See [Adding a New User](#), on page 111 and [Changing Your Password](#), on page 114.

Related Topics

[Configuring Users](#), on page 109

Configuring Password Options

If you chose to generate passwords for users automatically, they are configured in the following steps.

SUMMARY STEPS

1. Choose **Configure** > **User Defaults**.
2. Configure password options by performing the following tasks in the Password columns:
3. Click **Apply**.

DETAILED STEPS

Step 1 Choose **Configure** > **User Defaults**.

The system displays the Configure User Defaults page.

Step 2 Configure password options by performing the following tasks in the Password columns:

Note Although there is space to set a PIN, the Cisco Unified SIP Proxy system does not use PINs. If you set values here, they will not be used.

- a) Select whether the auto-generation policy will be **random** or **blank**.
- b) (Optional) Check **Enable expiry (days)** to set an expiration date for the password. The range is 3 to 365.
- c) Set the history depth. The range is 1 to 10.
- d) Select the minimum length of the password. The range for the password is 8 to 64.

Step 3 Click **Apply**.

Configuring Account Lockout Policy

The account lockout policy determines how the system acts when a user tries to log in and fails.

SUMMARY STEPS

1. Choose **Configure > User Defaults**.
2. Choose one of the following lockout policy types for the Password field:
3. Click **Apply** to save your settings.

DETAILED STEPS

Step 1 Choose **Configure > User Defaults**.

The system displays the Configure User Defaults page.

Step 2 Choose one of the following lockout policy types for the Password field:

Note Although there is space to set a PIN, the Cisco Unified SIP Proxy system does not use PINs. If you set values here, they will not be used.

- **Disable lockout**—The user can continue to try to login with no consequences for failing.
- **Permanent**—The user is permanently locked out after a certain number of failed login attempts. Enter the maximum number of failed attempts. The range is 1 to 200.
- **Temporary**—The user is temporarily locked out of the system. Enter values for the following:
 - Number of allowable attempts. The range is 1 to 200.
 - Temporary lockout duration. Pick any number in minutes.
 - Maximum number of failed attempts. The range is 1 to 200.

Step 3 Click **Apply** to save your settings.



CHAPTER 18

Configuring Groups

- [Viewing a List of Groups, on page 117](#)
- [Adding a New User Group, on page 118](#)
- [Subscribing Members or Owners to a Group, on page 119](#)
- [Unsubscribing Members and Owners from a Group, on page 120](#)
- [Displaying or Modifying Group Parameters, on page 121](#)
- [Viewing Owners and Members of a Group, on page 122](#)
- [Modifying Group Ownership and Membership in Other Groups, on page 122](#)
- [Deleting a Group, on page 123](#)
- [Finding a Group, on page 124](#)
- [About Capabilities, on page 124](#)

Viewing a List of Groups



Important

You cannot modify the default group and associated capabilities. You can only associate the non-default group with one or more newly created privileges.

You cannot associate the default or non-default group as a member of a non-default group.

You can delete the non-default group only on unsubscribing all the users who are subscribed as its members.

Step 1

Choose **Configure > Groups**.

The system displays the Configure Groups page, containing the following fields:

- Group ID
- Display Name
- Primary Extension
- Privileges

Step 2

To see a different number of groups on each page, choose another number from the drop-down box on the top right and click **Go**. You can choose to see 10, 25, 50, 100, or all groups.

- Step 3** To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
- Step 4** To sort groups, click any of the headers.

Group Fields

The table lists the fields on the page.

Table 27: Group Parameters

| Parameter | Description |
|----------------------|--|
| Group ID | Alphanumeric user identifier. |
| Full name | Long name of the group as it should appear on telephone displays. |
| Description | Description of the group. The word “group” is automatically added to the Group ID entry. |
| Primary Extension | Primary extension of the group. |
| Primary E.164 Number | Associates a full telephone number and area code with this group. |
| Fax Number | Associates a fax number with this group. |

Related Topics

[Managing the System Configuration](#), on page 173

Adding a New User Group

Before you begin

- Configuring one or more groups is optional.
- Determine the primary extension to be assigned to the group. Ensure that this extension is active.

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Click **Add**.
3. Enter information into the fields shown below:
4. Check the check box next to the capabilities for this group to have. See [About Capabilities, on page 124](#).
5. Click **Add**.

DETAILED STEPS

- Step 1** Choose **Configure > Groups**.
The system displays the Configure Groups page.
- Step 2** Click **Add**.
The system displays the Add a New Group page.
- Step 3** Enter information into the fields shown below:
- Group ID
 - Full name
 - Description—The word “group” is automatically added to the Group ID entry. You can add more text to this description.
 - Primary Extension for the group
 - Primary E.164 Number
 - Fax Number
- Step 4** Check the check box next to the capabilities for this group to have. See [About Capabilities, on page 124](#).
- Step 5** Click **Add**.
The system displays the Configure Groups page, with the new group in the table.
-

Subscribing Members or Owners to a Group

To subscribe members or owners of a group, complete the following steps:

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Click the underlined name of the group to which you are adding new members or owners.
3. Click the **Owners/Members** tab.
4. To add a new member, click **Subscribe Member**. To add a new owner, click **Subscribe Owner**.
5. Under type, select either users or groups. Enter the user ID or Group ID, name or description, or the extension of the person or group to add to this group.
6. Click **> Find**.
7. Do one of the following:
8. To add more members or owners to the group, repeat **Step 4** through **Step 7**.

DETAILED STEPS

- Step 1** Choose **Configure > Groups**.

The system displays the Configure Groups page.

Step 2 Click the underlined name of the group to which you are adding new members or owners.

The system displays the Group Profile page for that group.

Step 3 Click the **Owners/Members** tab.

The system displays all owners and members of the group.

Step 4 To add a new member, click **Subscribe Member**. To add a new owner, click **Subscribe Owner**.

The system displays the Find page.

Step 5 Under type, select either users or groups. Enter the user ID or Group ID, name or description, or the extension of the person or group to add to this group.

Step 6 Click **> Find**.

The system displays all users or groups that meet the search criteria.

Step 7 Do one of the following:

- Add one or more member or owner to the group by checking the check box next to each selected member's or owner's name and clicking **Select Rows**. The system displays the Group page with the new member or owner added.
- Look for other people to add by clicking **Back to Find** without checking a check box next to any name. The system displays the Find page. Return to **Step 5** and continue.

Step 8 To add more members or owners to the group, repeat **Step 4** through **Step 7**.

Unsubscribing Members and Owners from a Group



Restriction Only group owners can delete members and owners.

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Click the underlined name of the group to manage.
3. Click the **Owners/Members** tab.
4. Check the check box next to the name of each member or owner who you want to unsubscribe from this group.
5. Click **Unsubscribe**.

DETAILED STEPS

Step 1 Choose **Configure > Groups**.

The system displays the Configure Groups page.

- Step 2** Click the underlined name of the group to manage.
The system displays the Group Profile page for this group.
- Step 3** Click the **Owners/Members** tab.
The system displays all owners and members of the group.
- Step 4** Check the check box next to the name of each member or owner who you want to unsubscribe from this group.
- Step 5** Click **Unsubscribe**.
The system displays the Group Members page with the members or owners removed.
-

Displaying or Modifying Group Parameters

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Click the underlined name of the group to view or modify.
3. To edit these fields, enter the new information and click **SaveApply**.

DETAILED STEPS

- Step 1** Choose **Configure > Groups**.
The system displays the Configure Groups page.
- Step 2** Click the underlined name of the group to view or modify.
The system displays the Group Profile page for this group, with the following fields:
- Group ID
 - Full name
 - Description
 - Primary Extension
 - Primary E.164 number
 - Fax Number
 - Capabilities. See [About Capabilities, on page 124](#).
- Step 3** To edit these fields, enter the new information and click **SaveApply**.
-

Viewing Owners and Members of a Group

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Click the underlined name of the group to view.
3. Click the **Owners/Members** tab to see the users who are owners or members of this group.
4. Click any column heading to sort by that subject.

DETAILED STEPS

-
- Step 1** Choose **Configure > Groups**.
The system displays the Configure Groups page.
- Step 2** Click the underlined name of the group to view.
The system displays the Group Profile page for that group.
- Step 3** Click the **Owners/Members** tab to see the users who are owners or members of this group.
The system displays the Owners/Members page.
- Step 4** Click any column heading to sort by that subject.
-

Modifying Group Ownership and Membership in Other Groups

A group has its own set of members, but a group can also be assigned as a member or an owner of one or more other groups. If a group is assigned as an owner of another group, any individual member of the owner group has privileges as an owner of the owned group. For example, if the Administrator group is added as an owner of the Technical Support group, any individual member of the Administrator group can add, modify, or delete members of the Technical Support group. Additionally, individual users that do not belong to another group can be added as owners of the Technical Support group.

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Click the name of the group whose membership you want to modify.
3. Click the **Owner/Member of Groups** tab.
4. To see a different number of groups on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or 500 groups.
5. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
6. To sort groups, click any of the headers.
7. To designate your group as an owner of another group, click **Subscribe as owner**. To subscribe your group as a member of another group, click **Subscribe as member**.

8. Enter the group ID, description, or extension of the groups to find.
9. Click **Find**.
10. To select one or more groups, check the check box next to each group's name and click **Select Rows**.

DETAILED STEPS

- Step 1** Choose **Configure > Groups**.
The system displays the Configure Groups page.
- Step 2** Click the name of the group whose membership you want to modify.
The system displays the Group Profile page for that group.
- Step 3** Click the **Owner/Member of Groups** tab.
The system displays the Owner/Member of Groups page.
- Step 4** To see a different number of groups on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or 500 groups.
- Step 5** To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
- Step 6** To sort groups, click any of the headers.
- Step 7** To designate your group as an owner of another group, click **Subscribe as owner**. To subscribe your group as a member of another group, click **Subscribe as member**.
The system displays the Find page.
- Step 8** Enter the group ID, description, or extension of the groups to find.
- Step 9** Click **Find**.
The system displays all the groups that meet the search criteria.
- Step 10** To select one or more groups, check the check box next to each group's name and click **Select Rows**.
The system adds the new groups to the list of groups on the Owner/Member of Groups page.
-

Deleting a Group

Deleting a group does not delete the members of the group.

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Check the check box next to the name of the group to delete.
3. Click **Delete**.
4. At the prompt, click **OK** to delete the group.

DETAILED STEPS

- Step 1** Choose **Configure > Groups**.
The system displays the Configure Groups page.
- Step 2** Check the check box next to the name of the group to delete.
- Step 3** Click **Delete**.
- Step 4** At the prompt, click **OK** to delete the group.
-

Finding a Group

Use this procedure to search for a group.

SUMMARY STEPS

1. Choose **Configure > Groups**.
2. Click **Find**. The following fields appear in the Find Groups window:
3. Enter the search criteria in one or more fields and click **Find**.

DETAILED STEPS

- Step 1** Choose **Configure > Groups**.
The system displays the Configure Groups page.
- Step 2** Click **Find**. The following fields appear in the Find Groups window:
- Group ID
 - Description
 - Extension
- Step 3** Enter the search criteria in one or more fields and click **Find**.
The system displays the Configure Groups page with the results of your search.
-

About Capabilities

You can assign capabilities to groups. Cisco Unified SIP Proxy has three capabilities:

- pfsread—Allows users to read from the public file system (PFS).
- pfsreadwrite—Allows users to read from and write to the PFS.

- superuser—Gives administrator privileges to users in this group.



CHAPTER 19

Configuring Privileges

- [Viewing Privileges, on page 127](#)
- [Creating a Privilege, on page 130](#)
- [Editing a Privilege, on page 130](#)

Viewing Privileges

SUMMARY STEPS

1. Choose **Configure > Privileges**.
2. To see a different number of privileges on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all privileges.
3. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
4. To sort the privileges, click any header.
5. To delete a privilege, do the following:

DETAILED STEPS

- Step 1** Choose **Configure > Privileges**.
The system displays the Configure Privileges page.
- Step 2** To see a different number of privileges on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all privileges.
- Step 3** To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
- Step 4** To sort the privileges, click any header.
- Step 5** To delete a privilege, do the following:
- a) Select the privilege to delete.
 - b) Click **Delete**.

Tip You cannot delete the pfsread, pfsreadwrite, or the superuser privileges. However, privileges that are linked to a group can be deleted without prior warning and this will result in the group not having any privileges.

Overview of Privileges

Cisco Unified SIP Proxy provides three predefined privileges that you can assign to groups. You can also create your own privileges and modify the predefined privileges.

When you assign a privilege to a group, any member of the group is granted the privilege rights. An administrator group is created automatically by the software initialization process from the imported subscribers designated as administrators.

When you create or modify privileges, you add or delete the operations allowed by that privilege. Operations define the CLI commands and GUI functions that are allowed. Most operations include only one CLI command and GUI function. In addition to adding operations to a privilege, you can also configure a privilege to have another privilege nested inside of it. A privilege configured with a nested privilege includes all operations configured for the nested privilege.

[List of Operations, on page 128](#) describes all available operations that you can add to privileges.



Note You cannot modify the superuser privilege. The superuser privilege includes all the operations.

To configure privileges, see [Creating a Privilege, on page 130](#).

Related Topics

[Viewing Privileges, on page 127](#)

List of Operations

Table 28: List of Operations

| Operation | Description |
|----------------------------|---|
| culp.configuration | Configure culp read and write access. |
| culp.readonlyconfiguration | Configure culp readonly access. |
| group.configuration | Create, modify, and delete groups. |
| security.aaa | Configure and modify AAA service settings. |
| security.access | Configure system level security regarding encryption of data, including defining crypto keys. Note Also includes permission to reload the system. |

| Operation | Description |
|------------------------|--|
| security.password | Configure settings for the system password and policy, such as: <ul style="list-style-type: none"> • Expiry • Lockout (temporary and permanent) • History • Length |
| security.pin | Configure settings for the system PIN and policy, such as: <ul style="list-style-type: none"> • Expiry • Lockout (temporary and permanent) • History • Length |
| services.configuration | Configure system services: DNS, NTP/clock, SMTP, SNMP, Fax Gateway, Cisco UMG, hostname, domain, interfaces (counters), and system default language. <p>Note Also includes permission to reload the system.</p> |
| services.manage | System level services commands not related to configuration like clearing DNS cache and ping. |
| software.install | Install, upgrade, or inspect system software or add-ons such as languages and licenses. <p>Note Also includes permission to reload the system.</p> |
| system.backup | Configure backup. |
| system.configuration | Configure system settings such as the clock, hostname, domain name, default language, and interfaces (counters). |
| system.debug | Collect and configure trace and debug data. Includes copying data like core and log files. |
| system.view | View system settings and configuration. |

| Operation | Description |
|--------------------|--|
| user.configuration | Create, modify, and delete users and groups, including the configuration of: <ul style="list-style-type: none"> • First and Last Name • Nickname • Display Name • Language |
| user.password | Create, set, or remove others passwords. |
| user.pin | Create, set, or remove others PINs. |

Related Topics

[Configuring Privileges](#), on page 127

Creating a Privilege

SUMMARY STEPS

1. Choose **Configure > Privileges**.
2. Click **Add**.
3. Enter a name and description for the privilege.
4. Check the operations to add to the privilege. See [List of Operations, on page 128](#).
5. Click **Add**.

DETAILED STEPS

-
- Step 1** Choose **Configure > Privileges**.
The system displays the Configure Privileges page.
- Step 2** Click **Add**.
- Step 3** Enter a name and description for the privilege.
- Step 4** Check the operations to add to the privilege. See [List of Operations, on page 128](#).
- Step 5** Click **Add**.
-

Editing a Privilege

Before you begin

- You cannot modify the pfsread, pfsreadwrite, or the superuser privilege.

- Some operations are mandatory and cannot be removed.
- Create a privilege. See [Creating a Privilege, on page 130](#).

SUMMARY STEPS

1. Choose **Configure > Privileges**.
2. Click the underlined name of the privilege to customize.
3. Select the operations to add to the privilege or deselect the operations to remove.
4. Click **Apply**.
5. Click **OK** to save your changes.

DETAILED STEPS

-
- Step 1** Choose **Configure > Privileges**.
The system displays the Configure Privileges page.
- Step 2** Click the underlined name of the privilege to customize.
- Step 3** Select the operations to add to the privilege or deselect the operations to remove.
- Step 4** Click **Apply**.
- Step 5** Click **OK** to save your changes.
-



CHAPTER 20

Configuring Authentication, Authorization, and Accounting

- [Configuring the AAA Authentication Server, on page 133](#)
- [Specifying the Policy that Controls the Behavior of Authentication and Authorization, on page 136](#)
- [Configure AAA Accounting Server, on page 136](#)

Configuring the AAA Authentication Server

The two procedures for configuring AAA authentication consist of:

- Configuring connection parameters for the AAA authentication server
- Configuring whether the authentication servers or local authentication database will be queried first



Note To help protect the cryptographic information of the RADIUS server, you must view the running configuration to see this information.

- [About the Authentication Order, on page 133](#)
 - [About Authentication Failover, on page 134](#)
 - [About Unreachable Failover, on page 134](#)
 - [Example of Authentication Sequence, on page 135](#)
 - [Configuring Connection Parameters for the AAA Authentication Server, on page 135](#)
-

About the Authentication Order

The AAA policy specifies the failover functionality that you can optionally configure for the authentication server. You can use these two types of failover functionality separately or in combination:

- Authentication failover
- Unreachable failover

About Authentication Failover

The authentication failover feature enables you to optionally use a remote RADIUS server for user login authentication, in addition to the local database. The procedure in this section configures the order in which authentication is resolved. You can configure authentication to use:

- The local database only
- The remote server only
- The local database first, then the remote server
- The remote server first, then the local database

When using both local and remote authentication, you can also configure whether you want the user attributes that are retrieved from a remote RADIUS AAA server to be merged with the attributes found in the local user database for the same username.



Note When using AAA authentication, a user configured only on the remote radius server (and not on the local Cisco Unified SIP Proxy user database) will have low privilege levels and limited GUI access upon logging into Cisco Unified SIP Proxy. To enable higher privilege levels for this user, configure a local user with the same username as that on the Radius server, and assign the appropriate authorization levels. For detailed information, see the [Application Note on AAA based authentication](#).

The authentication failover feature has the following limitations:

- Authentication with a RADIUS server is available only when accessing the GUI or CLI interface and requires only a user ID and password.
- Login information is not synchronized between the local system and the remote server. Therefore:
 - Any security features such, as password expiration, must be configured separately for Cisco Unified SIP Proxy and the RADIUS server.
 - Cisco Unified SIP Proxy users are not prompted when security events, such as password expiration or account lockout, occur on the RADIUS server.
 - RADIUS server users are not prompted when security events, such as password expiration or account lockout, occur on Cisco Unified SIP Proxy.

About Unreachable Failover

The Unreachable Failover feature is used only with RADIUS servers. This feature enables you to configure up to two addresses that can be used to access RADIUS servers.

As Cisco Unified SIP Proxy attempts to authenticate a user with the RADIUS servers, the system sends messages to users to notify them when a RADIUS server either cannot be reached or fails to authenticate the user.

Example of Authentication Sequence

In this example, authentication is performed by the remote server first, then by the local database. Also, two addresses are configured for the remote RADIUS server.

This sequence of events could occur during authentication for this example:

1. Cisco Unified SIP Proxy tries to contact the first remote RADIUS server.
2. If the first RADIUS server does not respond or does not accept the authentication credentials of the user, Cisco Unified SIP Proxy tries to contact the second remote RADIUS server.
3. If the second RADIUS server does not respond or does not accept the authentication credentials of the user, the user receives the appropriate error message and Cisco Unified SIP Proxy tries to contact the local database.
4. If the local database does not accept the authentication credentials of the user, the user receives an error message.

Configuring Connection Parameters for the AAA Authentication Server

SUMMARY STEPS

1. Choose **Configure > AAA > Authentication**.
2. Enter the following information in the appropriate fields for the primary server, and optionally, for the secondary server:
3. Click **Apply**.
4. Click **OK** to save your changes.

DETAILED STEPS

Step 1 Choose **Configure > AAA > Authentication**.

The system displays the AAA Authentication Server Configuration page.

Step 2 Enter the following information in the appropriate fields for the primary server, and optionally, for the secondary server:

- Server IP address or DNS name
- Port number used
- Cryptographic shared secret and security credentials
- Authentication order
- Number of login retries
- Length of login timeout
- Hostname
- Port
- Password

- Step 3** Click **Apply**.
- Step 4** Click **OK** to save your changes.
-

Specifying the Policy that Controls the Behavior of Authentication and Authorization

Use this procedure to configure the information used to log into the authentication server.

SUMMARY STEPS

1. Choose **Configure > AAA > Authorization**.
2. Select or deselect whether you want to merge the attributes of the remote AAA server with the attributes in the local database.
3. Click **Apply**.
4. Click **OK** to save your changes.

DETAILED STEPS

- Step 1** Choose **Configure > AAA > Authorization**.
- The system displays the Configure AAA Authorization Server Configuration window page.
- Step 2** Select or deselect whether you want to merge the attributes of the remote AAA server with the attributes in the local database.
- Step 3** Click **Apply**.
- Step 4** Click **OK** to save your changes.
-

Configure AAA Accounting Server

You can configure up to two AAA accounting servers. Automatic failover functionality is provided if you have two accounting servers configured. If the first server is unreachable, the accounting information is sent to the second server. If both accounting servers are unreachable, accounting records are cached until a server becomes available. If a server cannot be reached before the cache is full, the oldest accounting packets are dropped to make room for the new packets.

Because the configuration of the AAA accounting server is completely independent of the AAA authentication server, you can configure the AAA accounting server to be on the same or different machine from the AAA authentication server.

If you use a syslog server, it is not affected by the AAA configuration and continues to use the existing user interfaces. When the RADIUS server sends AAA accounting information to a syslog server, it is normalized into a single string before being recorded. If no syslog server is defined, the AAA accounting logs are recorded by the syslog server running locally on Cisco Unified SIP Proxy.



Note Only RADIUS servers are supported.

AAA Accounting Event Logging

AAA accounting logs contain information that enables you to easily:

- Audit configuration changes.
- Maintain security.
- Accurately allocate resources.
- Determine who should be billed for the use of resources.

You can configure AAA accounting to log the following types of events:

- Logins—All forms of system access, including access to the CLI and GUI, when a login is required.
- Logouts—All forms of system access, including access to the CLI and GUI, when a login is required before logout.
- Failed logins—Failed login attempts for all forms of system access, including access to the CLI and GUI, when a login is required.
- Configuration mode commands—Any changes made to the Cisco Unified SIP Proxy configuration using any interface such as CLI and GUI.
- EXEC mode commands—Any commands entered in Cisco Unified SIP Proxy EXEC mode using any interface such as CLI and GUI.
- System startups—System startups, which include information about the system's software version, installed licenses, installed packages, installed languages, and so on.
- System shutdowns—System shutdowns, which include information about the system's software version, installed licenses, installed packages, installed languages, and so on.

| Log Name | Description |
|-----------------|--|
| login | All forms of system access when a login is required. |
| logout | All forms of system access when a login is required before logout. |
| login-fail | Failed login attempts for all forms of system access when a login is required. |
| config-commands | Any changes made to the system configuration using any interface. |
| exec-commands | Any commands entered in EXEC mode using any interface. |

| Log Name | Description |
|-----------------|--|
| system-startup | System startups, which include information about the system's software version, installed licenses, installed packages, installed languages, and so on. |
| system-shutdown | System shutdowns, which include information about the system's software version, installed licenses, installed packages, installed languages, and so on. |

In addition to information specific to the type of action performed, the accounting logs also indicate the following:

- User that authored the action
- Time when the action was executed
- Time when the accounting record was sent to the server



Note Account logging is not performed during the system power-up playback of the startup configuration. When the system boots up, the startup-config commands are not recorded.

Configuring the AAA Accounting Server

Use this procedure to configure the information used to log into the accounting server.

SUMMARY STEPS

1. Choose **Configure > AAA > Accounting**.
2. Click **Accounting Enabled**.
3. Enter the following information in the appropriate field for the primary server, and optionally, for the secondary server:
4. Click **Apply**.
5. Click **OK** to save your changes.

DETAILED STEPS

-
- Step 1** Choose **Configure > AAA > Accounting**.
The AAA Accounting Server Configuration window appears.
- Step 2** Click **Accounting Enabled**.
- Step 3** Enter the following information in the appropriate field for the primary server, and optionally, for the secondary server:
- Server IP address or DNS name
 - Port number used

- Cryptographic shared secret and security credentials
- Number of login retries
- Length of login timeout

Step 4 Click **Apply**.

Step 5 Click **OK** to save your changes.

Configuring Accounting Event Logging

Use this procedure to configure which event types to log for AAA accounting.

SUMMARY STEPS

1. Choose **Configure > AAA > Accounting**.
2. Select the log events that you want to include in the log and deselect those you do not want to include.
3. Click **Apply** to save your changes.

DETAILED STEPS

Step 1 Choose **Configure > AAA > Accounting**.

The system displays the Accounting Server Configuration window.

Step 2 Select the log events that you want to include in the log and deselect those you do not want to include.

Step 3 Click **Apply** to save your changes.

Configuring the AAA Accounting Server and Event Logging

Use this procedure to configure the information used to log into the accounting server.

SUMMARY STEPS

1. Choose **Configure > AAA > Accounting**.
2. Enter the following information in the appropriate fields:
3. Select the log events to include in the log and deselect those to not include.
4. Click **Apply**.
5. Click **OK** to save your changes.

DETAILED STEPS

Step 1 Choose **Configure > AAA > Accounting**.

The system displays the Configure AAA Accounting page.

Step 2 Enter the following information in the appropriate fields:

- If accounting is enabled
- Number of login retries
- Length of login timeout, in seconds
- Server IP address or DNS name for the primary server
- Port number used for the primary server
- Password for the primary server
- Server IP address or DNS name for the secondary server
- Port number used for the secondary server
- Password for the secondary server

Step 3 Select the log events to include in the log and deselect those to not include.

Step 4 Click **Apply**.

Step 5 Click **OK** to save your changes.



CHAPTER 21

Viewing System Information

- [System Information, on page 141](#)

System Information

The system displays the System Information page with the following information:

| Parameter | Description |
|-----------------------|---|
| Module SKU | Unique ordering identifier for a Cisco Unified SIP Proxy. |
| Module Serial Number | Serial number of the Cisco Unified SIP Proxy. |
| Chassis Type | Not applicable |
| Chassis Serial Number | Not applicable |
| Software Version | Version of Cisco Unified SIP Proxy software that is running on this system. |
| Uptime | Amount of time that the Cisco Unified SIP Proxy has been running. |



CHAPTER 22

Configuring Domain Name Settings

- [Changing a DNS Server, on page 143](#)
- [Adding a DNS Server, on page 144](#)
- [Removing a DNS Server, on page 145](#)

Changing a DNS Server

Use this procedure to change the DNS servers if their names or IP addresses have changed.

Before you begin

Gather the following information:

- The hostname of the Cisco Unified SIP Proxy system.
- The domain name and IP address of the DNS server.

SUMMARY STEPS

1. Choose **System > Domain Name Settings**.
2. Change hostname or domain name of the server that stores the application files.
3. Click **Apply**.

DETAILED STEPS

Step 1 Choose **System > Domain Name Settings**.
The system displays the Domain Name Settings page.

Step 2 Change hostname or domain name of the server that stores the application files.

Step 3 Click **Apply**.

What to do next

Save the configuration. See [Managing the System Configuration, on page 173](#).

Reload the configuration. See [Using the Administration Control Panel, on page 171](#).

Related Topics

[Using the Administration Control Panel, on page 171](#)

[Managing the System Configuration, on page 173](#)

Adding a DNS Server

Enter additional DNS servers as alternate server destinations, to be used if the system cannot access the primary domain name server.



Restriction You can have a maximum of four DNS servers.

SUMMARY STEPS

1. Choose **System > Domain Name Settings**.
2. Click **Add** under Domain Name Service (DNS) Servers.
3. Enter the IP address of the server.
4. Click **Add**.

DETAILED STEPS

-
- Step 1** Choose **System > Domain Name Settings**.
The system displays the Domain Name Settings page.
- Step 2** Click **Add** under Domain Name Service (DNS) Servers.
The system displays the Add a DNS server page.
- Step 3** Enter the IP address of the server.
- Step 4** Click **Add**.
-

What to do next

Save the configuration. See [Managing the System Configuration, on page 173](#).

Reload the configuration. See [Using the Administration Control Panel, on page 171](#).

Related Topics

[Using the Administration Control Panel, on page 171](#)

[Managing the System Configuration, on page 173](#)

Removing a DNS Server

Use this procedure to delete a DNS server:

SUMMARY STEPS

1. Choose **System > Domain Name Settings**.
2. Check the check box next to the DNS server to delete.
3. Click **Delete**.
4. At the prompt, click **OK**.

DETAILED STEPS

- Step 1** Choose **System > Domain Name Settings**.
The system displays the Domain Name Settings page.
- Step 2** Check the check box next to the DNS server to delete.
- Step 3** Click **Delete**.
- Step 4** At the prompt, click **OK**.
-

What to do next

Save the configuration. See [Managing the System Configuration](#), on page 173.

Reload the configuration. See [Using the Administration Control Panel](#), on page 171.

Related Topics

[Using the Administration Control Panel](#), on page 171

[Managing the System Configuration](#), on page 173



CHAPTER 23

Configuring Network Time and Time Zone Settings

- [Adding NTP Server and Configuring Time Zone, on page 147](#)
- [Adding an NTP Server, on page 147](#)
- [Removing an NTP Server, on page 148](#)
- [Setting an NTP Server as the Preferred Server, on page 149](#)
- [Changing the Time Zone, on page 149](#)

Adding NTP Server and Configuring Time Zone

You must add an NTP server to your Cisco Unified SIP Proxy system and configure the time zone to ensure that system processes have the correct date and time associated with them.

Adding an NTP Server



Restriction You can have a maximum of three NTP servers.

SUMMARY STEPS

1. Choose **System > Network Time & Time Zone Settings**.
2. Click **Add**.
3. Enter the hostname or IP address of the NTP server. To make it the primary NTP server, check the **Preferred** check box.
4. Click **Add**.

DETAILED STEPS

- Step 1** Choose **System > Network Time & Time Zone Settings**.
The system displays the Network Time & Time Zone Settings page.

- Step 2** Click **Add**.
The system displays the Add a NTP Server page.
- Step 3** Enter the hostname or IP address of the NTP server. To make it the primary NTP server, check the **Preferred** check box.
- Step 4** Click **Add**.
The system displays the Network Time and Time Zone Settings page with the new server listed in the table.
-

What to do next

Save the configuration. See [Managing the System Configuration](#), on page 173.

Reload the configuration. See [Using the Administration Control Panel](#), on page 171.

Related Topics

[Using the Administration Control Panel](#), on page 171

[Managing the System Configuration](#), on page 173

Removing an NTP Server

SUMMARY STEPS

1. Choose **System > Network Time & Time Zone Settings**.
2. Check the check box next to the NTP server to remove.
3. Click **Delete**.
4. Click **OK** at the prompt.

DETAILED STEPS

-
- Step 1** Choose **System > Network Time & Time Zone Settings**.
The system displays the Network Time & Time Zone Settings page.
- Step 2** Check the check box next to the NTP server to remove.
- Step 3** Click **Delete**.
- Step 4** Click **OK** at the prompt.
-

What to do next

Save the configuration. See [Managing the System Configuration](#), on page 173.

Reload the configuration. See [Using the Administration Control Panel](#), on page 171.

Related Topics

[Using the Administration Control Panel](#), on page 171

[Managing the System Configuration](#), on page 173

Setting an NTP Server as the Preferred Server



Restriction You must have at least two NTP servers.

SUMMARY STEPS

1. Choose **System** > **Network Time & Time Zone Settings**.
2. Check the check box next to the NTP server to set as the preferred server.
3. Click **Preferred**.
4. Click **OK**.

DETAILED STEPS

- Step 1** Choose **System** > **Network Time & Time Zone Settings**.
The system displays the Network Time & Time Zone Settings page.
- Step 2** Check the check box next to the NTP server to set as the preferred server.
- Step 3** Click **Preferred**.
- Step 4** Click **OK**.

What to do next

Save the configuration. See [Managing the System Configuration](#), on page 173.

Reload the configuration. See [Using the Administration Control Panel](#), on page 171.

Related Topics

[Using the Administration Control Panel](#), on page 171

[Managing the System Configuration](#), on page 173

Changing the Time Zone

SUMMARY STEPS

1. Choose **System** > **Network Time & Time Zone Settings**.
2. Use the drop-down menu to select the correct country.
3. Use the drop-down menu to select the correct time zone.
4. Click **Apply**.
5. Click **OK** at the information prompt.

DETAILED STEPS

- Step 1** Choose **System > Network Time & Time Zone Settings**.
- The system displays the Network Time & Time Zone Settings page.
- Step 2** Use the drop-down menu to select the correct country.
- Step 3** Use the drop-down menu to select the correct time zone.
- Step 4** Click **Apply**.
- Step 5** Click **OK** at the information prompt.
-

What to do next

Save the configuration. See [Managing the System Configuration](#), on page 173.

Reload the configuration. See [Using the Administration Control Panel](#), on page 171.

Related Topics

[Using the Administration Control Panel](#), on page 171

[Managing the System Configuration](#), on page 173



CHAPTER 24

Configuring SNMP Settings

- [About SNMP, on page 151](#)
- [Adding, Editing, and Deleting an SNMP Community String, on page 151](#)
- [Adding, Editing, and Removing an SNMP Trap Host, on page 152](#)
- [Enabling SNMP Traps, on page 153](#)
- [Displaying MIBs, on page 153](#)
- [Editing the SNMPv2-MIB, on page 154](#)

About SNMP

Cisco Unified SIP Proxy supports SNMP MIBs and traps for monitoring its status. Cisco Unified SIP Proxy supports the following SNMP MIBs and traps:

- CISCO-USP-MIB
- CISCO-PROCESS-MIB

Adding, Editing, and Deleting an SNMP Community String

Use this procedure to add or edit an SNMP community. Communities can either be read-only or read-write only.



Restriction You can only define up to five read-only community strings and up to five read-write community strings.

SUMMARY STEPS

1. Choose **System > SNMP > Communities**.
2. To add an SNMP community string, do the following:
3. To edit an existing SNMP community string, do the following:
4. To remove an SNMP community string, do the following:

DETAILED STEPS

Step 1 Choose **System > SNMP > Communities**.

The system displays the SNMP Communities page.

Step 2 To add an SNMP community string, do the following:

a) In an empty space, enter the SNMP community string.

If there are no empty spaces, you must first delete another SNMP community string before you can add a new one. You can only define up to five read-only community strings and up to five read-write community strings.

b) Click **Update**.

Step 3 To edit an existing SNMP community string, do the following:

a) Go to the SNMP community string to edit and edit the name.

b) Click **Update**.

Step 4 To remove an SNMP community string, do the following:

a) Go to the SNMP community string to delete and highlight the name.

b) Click **Delete** on your keyboard.

c) Click **Update**.

Related Topics

[About SNMP](#), on page 151

Adding, Editing, and Removing an SNMP Trap Host

Configure an SNMP trap host to be notified of SNMP events. The system is configured to send all SNMP traps as they occur.



Restriction

- The hostname that you enter must be found in the DNS.
- You cannot edit the hostname after it has been entered and saved.

Before you begin

Gather the following information:

- The hostname of the SNMP trap host
- The community string of the SNMP trap host

SUMMARY STEPS

1. Choose **System > SNMP > Hosts**.
2. To add an SNMP trap host, do the following:
3. To edit an existing SNMP trap host, do the following:

4. To remove an SNMP trap host, do the following:

DETAILED STEPS

- Step 1** Choose **System > SNMP > Hosts**.
The system displays the SNMP Trap Hosts page.
- Step 2** To add an SNMP trap host, do the following:
- a) Click **Add**.
The system displays the SNMP Host Profile page.
 - b) Enter the hostname and the community string for the SNMP trap.
 - c) Click **Update**.
- Step 3** To edit an existing SNMP trap host, do the following:
- a) Click the underlined hostname of the SNMP trap host to edit.
The system displays the SNMP Host Profile page.
 - b) Edit the community string for the SNMP trap.
 - c) Click **Update**.
- Step 4** To remove an SNMP trap host, do the following:
- a) Check the check box next to the SNMP trap host.
 - b) Click **Remove**.

Related Topics

[Enabling SNMP Traps](#), on page 153

Enabling SNMP Traps

- Step 1** Choose **System > SNMP > Traps**.
The system displays the SNMP Trap page.
- Step 2** Select if you want to enable or disable SNMP traps.
Check **Select All** to select all traps available.
- Step 3** Click **Update**.
-

Displaying MIBs

Use this procedure to display a list of the MIBs for Cisco Unified SIP Proxy.

SUMMARY STEPS

1. Choose **System > SNMP > MIBs**.
2. To enable the traps for all the SNMP MIBs, check **Enable SNMP Traps** and click **Update**.

DETAILED STEPS

-
- Step 1** Choose **System > SNMP > MIBs**.
The system displays the SNMP MIBs page.
- Step 2** To enable the traps for all the SNMP MIBs, check **Enable SNMP Traps** and click **Update**.
-

Editing the SNMPv2-MIB



Restriction The only MIB that you can edit is the SNMPv2-MIB.

SUMMARY STEPS

1. Choose **System > SNMP > MIBs**.
2. Click the underlined name of the SNMPv2-MIB.
3. Enter or update the contact or location for the SNMPv2-MIB.
4. Click **Update**.

DETAILED STEPS

-
- Step 1** Choose **System > SNMP > MIBs**.
The system displays the SNMP MIBs page.
- Step 2** Click the underlined name of the SNMPv2-MIB.
The system displays the SNMPv2-MIB page.
- Step 3** Enter or update the contact or location for the SNMPv2-MIB.
- Step 4** Click **Update**.
-



CHAPTER 25

Configuring System Login Banner

Use this procedure to change the text on the login banner that users see when they log in to the CLI.

SUMMARY STEPS

1. Choose **System > Login Banner**.
2. Enter the text for the login banner.
3. Click **Apply** to save your settings.

DETAILED STEPS

-
- | | |
|---------------|--|
| Step 1 | Choose System > Login Banner . The system displays the Login Banner page. |
| Step 2 | Enter the text for the login banner. |
| Step 3 | Click Apply to save your settings. |
-



CHAPTER 26

Monitoring the Cisco Unified SIP Proxy System



Note In Cisco Unified SIP Proxy 10.2.2 and later versions, Calls Per Second and System Resources pages are removed from Cisco Unified SIP Proxy Graphical User Interface(GUI). Ensure to execute the following show Command-line Interface (CLI) to view the Calls Per Second and System Resources data:

- **show cpu-usage history**60-seconds | 60-minutes |72-hours . Refer show cpu-usage history chapter in [CLI Command Reference for Cisco Unified SIP Proxy Release 10.2](#)
- **show memory-usage history**60-seconds | 60-minutes |72-hours. Refer show memory-usage history chapter in [CLI Command Reference for Cisco Unified SIP Proxy Release 10.2](#)
- **show cps-history**60-minutes |72-hours . Refer show cps-history chapter in [CLI Command Reference for Cisco Unified SIP Proxy Release 10.2](#)

- [Monitoring the Call Statistics, on page 157](#)
- [Monitoring the Server Group Status, on page 158](#)

Monitoring the Call Statistics



Restriction The system only displays the Active Calls data if call admission control is enabled.

SUMMARY STEPS

1. Choose **Monitor > Calls Statistics**.
2. To reset the number of call to zeros, check either Total Calls or Active Calls (or both) and click **Reset**.

DETAILED STEPS

Step 1 Choose **Monitor > Calls Statistics**.

The system displays the Call Statistics page with two sections:

- The Total Calls section lists the total number of calls into the server and the number of failed calls.

- The Active Calls section lists the number of active calls and the number of calls that timed out.

Step 2 To reset the number of call to zeros, check either Total Calls or Active Calls (or both) and click **Reset**.

Monitoring the Server Group Status

Monitor the status of the server groups and elements to ensure that they do not stop working.



Tip If a server group or element goes down, check that SIP pingging is set up so that the proxy will know when the server group or element comes back up.

SUMMARY STEPS

1. Choose **Monitor > Server Group Status**. See [Server Group Status Page, on page 159](#).
2. To expand the lists, click **Expand All**. To condense the lists, click **Collapse All**.
3. To see statistics about a particular endpoint, click the underlined value under either Active Calls/Allowed Limit or Total Calls/Failures (% success). The system displays the Call Statistics page for that endpoint with the following information:

DETAILED STEPS

- Step 1** Choose **Monitor > Server Group Status**. See [Server Group Status Page, on page 159](#).
- Step 2** To expand the lists, click **Expand All**. To condense the lists, click **Collapse All**.
- Step 3** To see statistics about a particular endpoint, click the underlined value under either Active Calls/Allowed Limit or Total Calls/Failures (% success). The system displays the Call Statistics page for that endpoint with the following information:
- IP address
 - Port
 - Transport type
 - Network
 - Number of total calls
 - Number of failed calls
 - Success percentage
 - Number of active calls (only if call admission control is enabled)

You can reset some of these values by checking the check box and clicking **Reset**.

Server Group Status Page

The Server Group Status page that lists the following information:

Table 29: Status Page Information

| Field | Description |
|----------------------------|---|
| Server Group/Element | Displays the name of the SIP server group. |
| Status | Displays the operational status of the SIP server group. |
| Q-Value | <p>Displays a real number that indicates the priority of the server group element with respect to others in the server group.</p> <p>The Q-value provides the priority of each member (element) which varies from 0.0 to 1.0, where 1.0 is the highest priority.</p> <p>Note These values will be blank if there are multiple elements for the server group and the display is not expanded to show all elements.</p> |
| Weight | <p>Displays the percentage assigned to the request-URI or route-URI element in the route group if implementing weight-based routing.</p> <p>Note These values will be blank if there are multiple elements for the server group and the display is not expanded to show all elements.</p> |
| Active Calls/Allowed Limit | <p>Displays the following:</p> <ul style="list-style-type: none"> • number of active sessions • allowed limit <p>Note Only displays a value if the following criteria are met:</p> <ul style="list-style-type: none"> • call admission control is enabled; otherwise, it displays “N/A” • row contains an actual endpoint (as opposed to a top-level or nested server group; otherwise, the area is blank) |

| Field | Description |
|----------------------------------|---|
| Total Calls/Failures (success %) | Displays the following: <ul style="list-style-type: none">• total number of sessions handled• total number of failed sessions• success rate |



Note The system does not refresh the information on this page. If you want to see updated values, refresh your browser.



CHAPTER 27

Viewing Reports

- [Viewing the Backup History Report, on page 161](#)
- [Viewing the Restore History Report, on page 162](#)
- [Viewing the Network Time Protocol Report, on page 162](#)

Viewing the Backup History Report

SUMMARY STEPS

1. Choose **Reports > Backup History**.
2. To see a different number of backup reports on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all backup reports.
3. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
4. To sort backup reports, click any of the headers.

DETAILED STEPS

Step 1 Choose **Reports > Backup History**.

If there is any backup history to report, the Backup History report contains the following fields:

- ID—ID of the backup.
- Server URL—The server on which the backup history is stored.
- Backup Time and Date—Date and time when the system was last backed up.
- Version—The version of the Cisco Unified SIP Proxy software that is installed.
- Description—A description of the backup.
- Result—Status of the last backup procedure. Result shows Success or Fail.

Step 2 To see a different number of backup reports on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all backup reports.

- Step 3** To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
- Step 4** To sort backup reports, click any of the headers.
-

Viewing the Restore History Report

SUMMARY STEPS

1. Choose **Reports > Restore History**.
2. To see a different number of restore history reports on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all restore history reports.
3. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
4. To sort restore history reports, click any of the headers.

DETAILED STEPS

- Step 1** Choose **Reports > Restore History**.

If there is any restore history to report, the Restore History report contains the following fields:

- ID—ID of the restore.
- Server URL—The server on which the restore history is stored.
- Restore Time and Date—Date and time when the system was last backed up.
- Version—The version of the Cisco Unified SIP Proxy software that is installed.
- Result—Status of the last restore procedure. Result shows Success or Fail for the components that were restored.
- Use the dialog box to change the number of rows displayed per window.

- Step 2** To see a different number of restore history reports on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all restore history reports.
- Step 3** To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
- Step 4** To sort restore history reports, click any of the headers.
-

Viewing the Network Time Protocol Report

SUMMARY STEPS

1. Choose **Reports > Network Time Protocol**.

DETAILED STEPS

Choose **Reports > Network Time Protocol**.

The report contains the following fields:

- #—The prioritized number of the NTP server. The system attempts to synchronize its time starting with NTP server number one.
 - NTP Server—IP address or hostname of the NTP server.
 - Status—Indicates if the NTP server connected with the Cisco Unified SIP Proxy or if it was rejected.
 - Time Difference (secs)—Time offset between the NTP server and the client.
 - Time Jitter (secs)—Estimated time error of the system clock, measured as an exponential average of RMS time differences.
-



CHAPTER 28

Configuring Backup and Restore

- [Configuring the Backup Server, on page 165](#)
- [Viewing Scheduled Backups, on page 167](#)
- [Adding a Scheduled Backup, on page 167](#)
- [Manually Starting a Backup, on page 169](#)
- [Starting a Restore, on page 169](#)

Configuring the Backup Server

Before you begin the backup process, set the backup configuration parameters. See [Backup Configuration Parameters, on page 165](#).

SUMMARY STEPS

1. Choose **Administration > Backup / Restore > Configuration**.
2. Enter the information shown in the following fields. See [Backup Configuration Parameters, on page 165](#).
3. Click **Apply** to save the information.

DETAILED STEPS

-
- Step 1** Choose **Administration > Backup / Restore > Configuration**.
The system displays the Backup / Restore Configuration page.
- Step 2** Enter the information shown in the following fields. See [Backup Configuration Parameters, on page 165](#).
- Step 3** Click **Apply** to save the information.
-

Backup Configuration Parameters

Gather the following values before you begin the backup process.



Note Cisco Unified SIP Proxy supports OpenSSH as the recommended SFTP server that can run in Linux, MacOS, or Windows systems. However, Cisco Unified SIP Proxy also supports the following Windows-based SFTP servers while configuring the backup folder:

Table 30: List of Cisco Unified SIP Proxy Supported SFTP Servers

| Configure Absolute Backup Path (/path/from/root/to/backup/directory) | Configure Relative Backup Path (/SFTP user landing path/to/backup/directory) |
|--|--|
| OpenSSH 8.0p1 | CopSSH 5.0.0 |
| GlobalScape 9.3.0 | FreeFTPd 1.0.13 |
| CoreFtp 1.2 | |
| Solarwinds 20.3.0.64 | |

Table 31: Backup Configuration Parameters

| Parameter | Description |
|-------------------|---|
| Server URL | <p>The absolute URL of the SFTP server on the network where the backup files are stored.</p> <p>Make sure that the SFTP server URL is pointing to the absolute path of the destination folder.</p> <p>The format should be <code>sftp://<server/directory>/</code> where <code><server/directory></code> is the IP address or hostname of the backup server followed by the absolute path of the directory.</p> <p>Note Ensure that you do not give any whitespaces in the backup path.</p> <p>To know if you need to provide the absolute path or relative path for the backup folder, see the <i>List of Cisco Unified SIP Proxy Supported SFTP Servers</i> table.</p> |
| User ID | <p>The user ID on the backup server.</p> <p>You must have an account on the server to which you are backing up your data. Do not use an anonymous user ID.</p> |
| Password | The password for the user ID on the backup server. |
| Maximum revisions | <p>The maximum number of revisions of the backup data to keep on the backup server.</p> <p>The maximum number is 50. The default value is 5.</p> |



Note Backing up and restoring data takes your Cisco Unified SIP Proxy to offline mode.

Viewing Scheduled Backups

SUMMARY STEPS

1. Choose **Administration > Backup / Restore > Scheduled Backups**.
2. To see a different number of scheduled backups on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all scheduled backups.
3. To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.
4. To sort scheduled backups, click any of the headers.

DETAILED STEPS

Step 1 Choose **Administration > Backup / Restore > Scheduled Backups**.

The system displays the Backup / Restore Scheduled Backups page with the following information:

- Name
- Description
- Schedule
- Next Run
- Categories of backup or type of data to save

Step 2 To see a different number of scheduled backups on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, 100, or all scheduled backups.

Step 3 To move to another page, use the left and right arrow buttons on the bottom right, or enter another page number and press **Enter**.

Step 4 To sort scheduled backups, click any of the headers.

Adding a Scheduled Backup

You can configure scheduled backups to occur once or recurring jobs that repeat:

- Every N days at a specific time
- Every N weeks on specific day and time
- Every N months on a specific day of the month and time

- Every N years on specific day and time

Before you begin

- Configure the server used to back up the data. See [Configuring the Backup Server, on page 165](#).
- Save your system configuration. See [Managing the System Configuration, on page 173](#).

SUMMARY STEPS

1. Choose **Administration > Backup / Restore > Scheduled Backups**.
2. Click **Schedule Backup**.
3. Enter a name and description for the scheduled backup.
4. Check the check box for the type of data to save. You can choose one or both:
5. From the Schedule tab, select the frequency of the scheduled backup:
6. Select whether the scheduled backup will start:
7. Click **Add**.

DETAILED STEPS

-
- Step 1** Choose **Administration > Backup / Restore > Scheduled Backups**.
The system displays the Backup / Restore Scheduled Backup page.
- Step 2** Click **Schedule Backup**.
The system displays the Backup / Restore Scheduled Backups page.
- Step 3** Enter a name and description for the scheduled backup.
- Step 4** Check the check box for the type of data to save. You can choose one or both:
- Configuration—Saves the configurations of the system and applications.
 - Data—Saves the routes and application data.
- Step 5** From the Schedule tab, select the frequency of the scheduled backup:
- Once
 - Daily
 - Weekly
 - Monthly
 - Yearly
- Step 6** Select whether the scheduled backup will start:
- Once
 - On a specific date and time

Step 7 Click **Add**.

Manually Starting a Backup

Before you begin

- Configure the server used to back up the data. See [Configuring the Backup Server, on page 165](#).
- Save your configuration. See [Managing the System Configuration, on page 173](#).

SUMMARY STEPS

1. Click **Administration > Backup / Restore > Start Backup**.
2. Enter a description of the backup file; for example, “backupdata6-2-04.”
3. Check the check box for the types of data to save. You can choose one or both:
4. Click **Start Backup**.
5. Click **OK** at the confirmation message.

DETAILED STEPS

Step 1 Click **Administration > Backup / Restore > Start Backup**.

The system displays the Backup / Restore Start Backup page and automatically generates a backup ID. The backup ID increases by one every time you back up the server.

Step 2 Enter a description of the backup file; for example, “backupdata6-2-04.”

Step 3 Check the check box for the types of data to save. You can choose one or both:

- Configuration—Saves the configurations of the system and applications.
- Data—Saves the routes and application data.

Step 4 Click **Start Backup**.

Step 5 Click **OK** at the confirmation message.

Starting a Restore

After you have backed up your configuration data, you can restore it for every new installation or upgrade.

Before you begin

Configure a backup server. See [Configuring the Backup Server, on page 165](#).

SUMMARY STEPS

1. Choose **Administration > Backup / Restore > Start Restore**.
2. Select the row containing the configuration to restore.
3. Check the check box for the type of data to save. You can choose one or both:

DETAILED STEPS

| | Command or Action | Purpose |
|---------------|---|--|
| Step 1 | Choose Administration > Backup / Restore > Start Restore . | <p>The system displays the Backup / Restore Start Restore page with the following fields:</p> <ul style="list-style-type: none"> • Backup ID —The backup ID of previous backups. • Version—Version • Description—Name of this backup. • Backup Time and Date—Date and time when this backup was made. • Categories—The type of data to restore. |
| Step 2 | Select the row containing the configuration to restore. | |
| Step 3 | Check the check box for the type of data to save. You can choose one or both: | <ul style="list-style-type: none"> • Configuration—Saves the configurations of the system and applications. • Data—Saves the routes and application data. |



CHAPTER 29

Using the Administration Control Panel

- [Reloading Cisco Unified SIP Proxy, on page 171](#)

Reloading Cisco Unified SIP Proxy



Restriction Reloading CUSP terminates all user sessions and lose all unsaved data.

SUMMARY STEPS

1. Choose **Administration > Control Panel**.
2. To reload CUSP, click **Reload Module**.
3. Click **OK** at the prompt.

DETAILED STEPS

-
- Step 1** Choose **Administration > Control Panel**.
The system displays the Control Panel page.
- Step 2** To reload CUSP, click **Reload Module**.
The system displays a dialog box warning you that reloading the system will lose any unsaved configuration data will be lost.
- Step 3** Click **OK** at the prompt.
-



CHAPTER 30

Managing the System Configuration

- [Restoring System Defaults, on page 173](#)
- [Viewing the Configuration Results, on page 174](#)
- [Previewing the Candidate Configuration, on page 174](#)

Restoring System Defaults

SUMMARY STEPS

1. Choose **Administration > Manage Configuration > Restore Defaults / Rollback**.
2. To save or commit the configuration, which makes this configuration the new starting configuration, do the following:
3. To restore the configuration to how it was when it was delivered from the factory, which means that you will lose all changes you have made and will reload the CUSP system, do the following:
4. To roll back the system to the most recent configuration, which replaces the current configuration and reloads the CUSP system, do the following:

DETAILED STEPS

-
- Step 1** Choose **Administration > Manage Configuration > Restore Defaults / Rollback**.
The system displays the Manage Configuration page.
- Step 2** To save or commit the configuration, which makes this configuration the new starting configuration, do the following:
- a) Click **Save/Commit Configuration**.
 - b) At the confirmation window, click **OK**.
- Step 3** To restore the configuration to how it was when it was delivered from the factory, which means that you will lose all changes you have made and will reload the CUSP system, do the following:
- a) Click **Restore Factory Defaults**.
 - b) At the confirmation window, click **OK**.
- Step 4** To roll back the system to the most recent configuration, which replaces the current configuration and reloads the CUSP system, do the following:
- a) Click **Rollback Active Configuration**.

- b) At the confirmation window, click **OK**.
-

Viewing the Configuration Results

After you save and commit the configuration, the system displays this web page that presents the result (either success or failure) of the save operation.

Previewing the Candidate Configuration

The system displays the code for the candidate configuration.



Note If there have not been any changes, the system displays the following message:

```
The candidate configuration contains no changes.
```

SUMMARY STEPS

1. Choose **Administration > Manage Configuration > Candidate Preview**.
2. To save or commit the configuration, which makes this configuration the new starting configuration, do the following:
3. To clear the system of the candidate configuration, which discards all uncommitted changes, do the following:

DETAILED STEPS

Step 1 Choose **Administration > Manage Configuration > Candidate Preview**.

The system displays the Candidate Configuration Preview page.

Step 2 To save or commit the configuration, which makes this configuration the new starting configuration, do the following:

- a) Click **Save/Commit Configuration**.
- b) At the confirmation window, click **OK**.

Step 3 To clear the system of the candidate configuration, which discards all uncommitted changes, do the following:

- a) Click **Clear Candidate Configuration**.
 - b) At the confirmation window, click **OK**.
-



CHAPTER 31

About Smart Licensing

Cisco Unified SIP Proxy supports smart licensing. In smart licensing, the purchased licenses are not tied to the hardware and Product Activation Key (PAK). Licenses can be configured by communication to the Smart Manager.

The smart licenses can be configured using the following procedures:

- [Configuring Smart License, on page 175](#)
- [Viewing the Smart License Summary, on page 176](#)

Configuring Smart License

SUMMARY STEPS

1. Launch Cisco Unified SIP Proxy GUI and choose **Administration > Smart License > Configuration**. The Smart Agent License page appears.
2. Click **Enable** radio button to configure smart licensing.
3. Enter the details in the fields. See [Smart Agent License Fields, on page 176](#) for field descriptions.
4. Check the **Enable Http(s)** check box.
5. Enter the proxy server address and port number in **Http(s) Proxy Address** field and **Port** fields.
6. Click **Update**.

DETAILED STEPS

-
- | | |
|---------------|---|
| Step 1 | Launch Cisco Unified SIP Proxy GUI and choose Administration > Smart License > Configuration . The Smart Agent License page appears. |
| Step 2 | Click Enable radio button to configure smart licensing. |
| Step 3 | Enter the details in the fields. See Smart Agent License Fields, on page 176 for field descriptions. |
| Step 4 | Check the Enable Http(s) check box. |
| Step 5 | Enter the proxy server address and port number in Http(s) Proxy Address field and Port fields. |
| Step 6 | Click Update . |
-

Smart Agent License Fields

Table 32: Smart Agent License Fields

| Parameter | Description |
|-------------------------------|---|
| Smart Agent Config | |
| License Count (multiple of 5) | Activates the requested number of licenses. The count should be multiple of 5. The count should be less than or equal to the maximum call rate that the Cisco Unified SIP Proxy can handle. |
| License Server url | Enter the Smart Manager server URL that connects to the central licensing server. Use the following URL for registering to cloud CSSM: https://tools.cisco.com/its/service/oddce/services/DDCEService . Use an appropriate URL for registering to an on-prem license server. |
| License Token ID | Specifies the token ID. It can be generated by the license server for the account that the Cisco Unified SIP Proxy instance is registered to. |
| Transport Mode | Specifies the protocol used to communicate with Smart Software Manager. Call home is the recommended Cisco proprietary secure protocol. HTTP is another optional protocol for communicating with Smart Software Manager. Note Smart Software Manager can be the cloud-based server or an on-premises license server. |
| Enable Http(s) Proxy | Enables the HTTP(S) proxy mode. You can use a web proxy to provide CUSP with access to CSSM over the Internet. |
| Http(s) Proxy Address | Sets the HTTP(S) proxy server address for accessing CSSM over the Internet. You can either use an IP address or FQDN. |

Viewing the Smart License Summary

The system displays the summary of the configured smart licenses.

Table 33: License Summary

| | |
|----------------------------|---|
| Smart License Client State | <p>Displays the state of the Smart Agent. The following is the list of states:</p> <ul style="list-style-type: none"> • Un-Configured—Smart licensing is not enabled. • Un-Identified—Smart licensing is enabled but the Smart Agent has not contacted Cisco Smart Software Manager (CSSM) to register. • Registered—The Smart Agent has contacted Cisco Smart Software Manager (CSSM) and registered. • Authorized—The Smart Agent enters Authorized state after registration when it receives a in compliance status in response to an entitlement authorization request to the Cisco Smart Software Manager (CSSM). • Out Of Compliance (OOC)—The Smart Agent enters out of compliance state after registration when it receives an Out of Compliance (OOC) status in response to an Entitlement Authorization request to the Cisco licensing authority. • Authorization Expired—If the device cannot communicate with Cisco for an extended period of time, usually 90 days, the agent goes into the Authorization expired state. |
| Product Serial Number | It's really a randomly generated unique virtual machine ID. |
| Product ID | Unique identifier for the Cisco Unified SIP Proxy. |
| License UDI | Combination of product ID and serial number generated randomly for identifying the Cisco Unified SIP Proxy. |
| License Server Address | Displays the address of the Smart Manager server provided while configuring. |
| HTTP Proxy Address | The proxy server address used, if configured, to reach the licensing server. |
| Smart Agent Transport Mode | Displays the protocol used to communicate with Smart Manager. |

| | |
|---------------------------|--|
| Licensing State | <p>Displays the licensing entitlement status of this instance. The following are the status:</p> <ul style="list-style-type: none"> • Eval—The Cisco Unified SIP Proxy is in Un-Identified state and the evaluation period has not expired. • InCompliance—The license count requested to the server is within the purchased limits. • OutOfcompliance—The license count requested is more than what is available in Cisco Smart Software Manager. • EvalExpried—Evaluation period of 90 days has expired. Calls are not allowed in EvalExpired state. • AuthorizationExpired—Authorization period has expired. Calls are not allowed in AuthorizationExpired state. |
| Product License Version | Displays the license version that the product instance is requesting. This is the same as the major software version. |
| Registration Expiry Date | Displays the expiry date and time when the license service identification certificate expires. Once expires, the device reverts to Un-Identified mode. |
| Next Auth Date | Displays the date and time for next license renewal. |
| Evaluation Period(in hrs) | Displays the number of hours left for Cisco Unified SIP Proxy to run on evaluation mode. The counter starts at 2160 (90 days) and counts down while in Un-Identified (unregistered) mode. This counter cannot be reset. |
| CPS Count Requested | Displays the number of calls per second licenses requested for. One license is used for every 5 calls per second requested. |
| Registration Successful | Identifies if registration was a success or failure. |
| Authorization Successful | Identifies if authorization was a success or failure. |
| Licensing Agent Status | Identifies if the Smart Agent is enabled or disabled. |
| Evaluation Mode | Identifies if the product is on evaluation mode. |
| Latest Failure Reason | Provides the reason due to which the latest license registration failed. |



CHAPTER 32

Manage Inactivity Timeout

You can increase the inactivity or idle timeout of your Cisco Unified SIP Proxy system to prevent logout of inactive sessions by setting the inactivity timer to an interval larger than the default interval duration.

- [Managing Inactivity Timeout, on page 179](#)

Managing Inactivity Timeout

SUMMARY STEPS

1. Choose **Administration > Manage Inactivity Timeout**.
2. Enter the inactivity timeout for your Cisco Unified SIP Proxy system.
3. Click **Set Inactivity Timeout** to save the information.

DETAILED STEPS

-
- Step 1** Choose **Administration > Manage Inactivity Timeout**.
The system displays the Manage Inactivity Timeout Configuration page.
- Step 2** Enter the inactivity timeout for your Cisco Unified SIP Proxy system.
You can set the value from 10 minutes to 24 hours.
- Step 3** Click **Set Inactivity Timeout** to save the information.
-



CHAPTER 33

Patch Upgrade

You can install a patch file that is downloaded from cisco.com on the existing Cisco Unified SIP Proxy release software. Installing the patch file helps you upgrade to a newer patch version of the Cisco Unified SIP Proxy without any modifications to your existing virtual machine.

- [Downloading the Patch File, on page 181](#)
- [Configuring Patch Upgrade, on page 182](#)
- [Installing the Patch File, on page 182](#)
- [Verifying Patch Upgrade, on page 183](#)
- [Troubleshooting Patch Upgrade, on page 184](#)

Downloading the Patch File

Cisco Unified SIP Proxy supports OpenSSH as the recommended SFTP server that can run in Linux, MacOS, or Windows systems. However, Cisco Unified SIP Proxy also supports the following Windows-based SFTP servers for storing the downloaded patch files:

Table 34: List of Cisco Unified SIP Proxy Supported SFTP Servers

| Configure Absolute Backup Path (/path/from/root/to/backup/directory) |
|--|
| Windows OpenSSH(recommended) |
| CopSSH-v5.0.0 |

- Step 1** Open the Cisco Unified SIP Proxy Server site: <http://www.cisco.com/c/en/us/support/unified-communications/unified-sip-proxy-software/tsd-products-support-series-home.html>.
- Step 2** If prompted, login using your Cisco.com username and password.
- Step 3** Locate the patch file in the “Download Software” section and download the file.

What to do next

Configure the patch upgrade. See [Configuring Patch Upgrade, on page 182](#).

Related Topics

[Configuring Patch Upgrade](#), on page 182

Configuring Patch Upgrade

Before you begin

- Know the SFTP server URL, username, and password. Make sure that the SFTP server URL is pointing to the absolute path of the directory containing the patch.
- Download the patch file and store it in the SFTP server. See [Downloading the Patch File](#), on page 181.

Step 1 Choose **Administration > Patch Upgrade**.

The system displays the Patch Upgrade Configuration page.

Step 2 Enter the URL of the Secured File Transfer Protocol server in the **SFTP Server URL** field.

Step 3 Enter the username in the **Username** field.

Step 4 Enter the password in the **Password** and **Confirm Password** fields.

Step 5 Click **Apply**.

What to do next

Install the patch file. See [Installing the Patch File](#), on page 182.

Related Topics

[Downloading the Patch File](#), on page 181

[Installing the Patch File](#), on page 182

Installing the Patch File

Before you begin

- Download the patch file from cisco.com and store it in the SFTP server. For detailed information on downloading the patch file, see [Downloading the Patch File](#), on page 181.
- Configure backup. For detailed information on configuring backup, see [Configuring Backup and Restore](#), on page 165.
- Complete patch upgrade configuration. For detailed information, see [Configuring Patch Upgrade](#).

Step 1 Choose **Administration > Start Upgrade**.

The system displays the Start Upgrade page.

Step 2 Select the patch file from the **File Name** drop-down list.

File Name drop-down lists all the patch files that are stored in the SFTP server. When there are no patch files stored in the SFTP server, the screen displays `No Patch file found in the server.` message.

Step 3 Click **OK** when prompted for confirmation.

Step 4 Click **Upgrade**.

The system goes offline and takes the backup of the existing configuration. The backup files are stored in the server specified while configuring backup and restore. After the backup is completed, the `Backup is done...` [Click here to start the patch installation](#) message is displayed.

Note Navigating to other pages terminates the upgrade process and requires you to restart the patch installation.

Step 5 Click **Click Here**.

- If installation is successful, the system gets upgraded to new version and displays `Patch upgrade successful. Bring back to online` message.
- If installation fails, the system rolls back to the version before initiating the patch install and displays `Patch upgrade failed. Bring back to online` message.

Step 6 Click **Bring Back to online**.

The system comes back to online mode.

What to do next

Verify the Cisco Unified SIP Proxy version. See [Verifying Patch Upgrade, on page 183](#).

Related Topics

[Configuring Patch Upgrade](#), on page 182

[Verifying Patch Upgrade](#), on page 183

Verifying Patch Upgrade

Step 1 Choose **Administration > Start Upgrade**.

The system displays the Start Upgrade page.

Step 2 Check the Cisco Unified SIP Proxy version displayed on the Start Upgrade page.

Related Topics

[Installing the Patch File](#), on page 182

[Troubleshooting Patch Upgrade](#), on page 184

Troubleshooting Patch Upgrade

Step 1 Choose **Administration > Start Upgrade**.

The system displays the Start Upgrade page.

Step 2 Click **Download Log File**. Save the file to a convenient location.

The log file in .txt format is downloaded. The log file is available in both success and failure scenarios. If there is an upgrade failure, you can use the log file for troubleshooting the failure.

Related Topics

[Verifying Patch Upgrade](#), on page 183



CHAPTER 34

Troubleshooting

- [Enabling Cisco Unified SIP Proxy Traces, on page 185](#)
- [Viewing the Cisco Unified SIP Proxy Log File, on page 187](#)
- [Configuring Trace Settings, on page 187](#)
- [Viewing Tech Support Information, on page 188](#)
- [Viewing a Trace Buffer, on page 189](#)
- [Viewing a Log File, on page 189](#)
- [Enabling SIP Message Logging, on page 190](#)
- [Searching SIP Message Calls, on page 191](#)
- [Viewing SIP Message Calls, on page 192](#)
- [Enabling the Failed Calls Log, on page 193](#)
- [Viewing the Failed Calls Log, on page 194](#)
- [Viewing the History of a Failed Call, on page 194](#)

Enabling Cisco Unified SIP Proxy Traces

SUMMARY STEPS

1. Choose **Troubleshoot** > **Cisco Unified SIP Proxy** > **Traces**.
2. To capture the network traffic on Cisco Unified SIP Proxy interfaces, check the **Packet Capture** check box.
3. To enable tracing on your system, check the **Enable Tracing** check box.
4. Set the trace values for the following components (For details on the level to choose for each component, see [Component Levels, on page 186](#)):
5. Click **Update** to save your changes.

DETAILED STEPS

Step 1 Choose **Troubleshoot** > **Cisco Unified SIP Proxy** > **Traces**.

The system displays the Cisco Unified SIP Proxy Traces page.

Step 2 To capture the network traffic on Cisco Unified SIP Proxy interfaces, check the **Packet Capture** check box.

a) Click **Start** to start packet capture.

b) Click **Stop** to stop packet capture.

Each packet capture request is limited to 40 MB. When the buffer size of the packet goes beyond 40 MB, the packet captures are overwritten, that is, the packet capture will always provide information of the last packet capture done. This prevents disk space over utilization. You can capture two packets of 20 MB each. This log file is located at: `/opt/CUSP/dsnrs/log/packetcapture`. The Cisco Unified SIP Proxy administrator must download the latest `packetcapture.zip` file before starting the next packet capture request.

Step 3 To enable tracing on your system, check the **Enable Tracing** check box.

Step 4 Set the trace values for the following components (For details on the level to chose for each component, see [Component Levels, on page 186](#)):

- Base Tracing
- Routing
- Proxy-Core
- SIP-Wire-Log
- Normalization
- Proxy-Transactions
- SIP-Ping
- License-Mgmt
- Trigger-Conditions
- Accounting
- SIP-Search
- Config-Mgmt

Step 5 Click **Update** to save your changes.

Related Topics

[Component Levels](#), on page 186

Component Levels

For each component, you can choose one of the following levels:

Component Levels

| Level | Description |
|---------|--|
| default | Uses the trace level of the parent. |
| debug | Logs messages of debug severity or higher. |
| info | Logs messages of info severity or higher. |
| warn | Logs messages of warning severity or higher. |

| Level | Description |
|-------|--|
| error | Logs messages of error severity or higher. |
| fatal | Logs messages of fatal severity or higher. |
| off | Does not log messages. |

Viewing the Cisco Unified SIP Proxy Log File

SUMMARY STEPS

1. Choose **Troubleshoot > Cisco Unified SIP Proxy > Log File**.
2. To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.
3. To save the trace log file information, do the following:

DETAILED STEPS

-
- Step 1** Choose **Troubleshoot > Cisco Unified SIP Proxy > Log File**.
The system displays the Cisco Unified SIP Proxy Trace Log File page and shows the contents of the trace log file.
- Step 2** To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.
- Step 3** To save the trace log file information, do the following:
- a) Click **Download Log File**.
 - b) Save the file to a convenient location.
 - c) Click **Close** when done.
-

Configuring Trace Settings

Use this procedure to enable traces, or debug message output, for components in the Cisco Unified SIP Proxy system. Components are entities and activities in the system. You can review the output by selecting **Troubleshoot > View > Trace Buffer**.



Restriction Enabling too many traces can adversely affect the system performance.

SUMMARY STEPS

1. Choose **Troubleshoot > Traces**.
2. To enable a trace on a system component, check the check box next to the name of the component.
3. Click **Apply** to save your changes.

4. Click **OK** in the confirmation window.

DETAILED STEPS

Step 1 Choose **Troubleshoot > Traces**.

The system displays the Traces page, with a hierarchical listing of the system components.

Step 2 To enable a trace on a system component, check the check box next to the name of the component.

- To expand the list of components, click the + sign next to any upper-level component. To condense the list of components, click the - sign next to any upper-level component.
- Check the check box next to any upper-level component to enable the traces for all of the components under that component. Uncheck the check box next to any upper-level component to disable the traces for all of the components under that component.

Step 3 Click **Apply** to save your changes.

Step 4 Click **OK** in the confirmation window.

Related Topics

[Viewing a Trace Buffer](#), on page 189

Viewing Tech Support Information

SUMMARY STEPS

1. Choose **Troubleshoot > View > Tech Support**.
2. To save the tech support information, click **Download Tech Support**.
3. Save the file to a convenient location.
4. Click **Close** when finished.

DETAILED STEPS

Step 1 Choose **Troubleshoot > View > Tech Support**.

The system displays the Tech Support page and shows a collection of configuration data.

Step 2 To save the tech support information, click **Download Tech Support**.

Step 3 Save the file to a convenient location.

Step 4 Click **Close** when finished.

Viewing a Trace Buffer

SUMMARY STEPS

1. Choose **Troubleshoot > View > Trace Buffer**.
2. To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.
3. To save the trace buffer information, do the following:
4. To clear the trace buffer, do the following:

DETAILED STEPS

-
- Step 1** Choose **Troubleshoot > View > Trace Buffer**.
- The system displays the Trace Buffer page and shows the contents of the trace buffer.
- Step 2** To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.
- Step 3** To save the trace buffer information, do the following:
- a) Click **Download Trace Buffer**.
 - b) Save the file to a convenient location.
 - c) Click **Close** when done.
- Step 4** To clear the trace buffer, do the following:
- a) Click **Clear Trace Buffer**.
 - b) Click **OK** at the confirmation prompt.
-

Viewing a Log File

SUMMARY STEPS

1. Choose **Troubleshoot > View > Log File**.
2. To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.
3. To save the log file, do the following:

DETAILED STEPS

-
- Step 1** Choose **Troubleshoot > View > Log File**.
- The system displays the Log File page and shows the contents of the log file.
- Step 2** To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.
- Step 3** To save the log file, do the following:

- a) Click **Download Log File**.
- b) Save the file to a convenient location.
- c) Click **Close** when done.

Enabling SIP Message Logging

Use the SIP message log to capture and troubleshoot SIP calls handled by Cisco Unified SIP Proxy. By default, the SIP message log is disabled. When the SIP message log is enabled, you can enter an optional expression to filter the messages that are stored.



Note If record-route is not configured for a network, the system does not display mid-dialog SIP messages in the SIP message log.



Caution Enabling the SIP message logging feature can have a significant performance impact on your system. We recommend that you limit the volume of calls processed by Cisco Unified SIP Proxy to less than 15 calls per second before you enable SIP message logging. We also recommend using the SIP message log filter whenever possible to limit the number of SIP messages that the system logs every second.

SUMMARY STEPS

1. Choose **Troubleshoot > SIP Message Log > Controls**.
2. Select if you want to enable or disable SIP message logging.
3. (Optional) Enter a regular expression filter. This reduces the number of calls that are written to the SIP message log. An example of a regular expression filter is **999...1020**. If you enter this, the system will match any number beginning with 999 and ending with 1020. Only messages that match the regular expression will pass through the filter and be stored.
4. Click **Update**.

DETAILED STEPS

-
- Step 1** Choose **Troubleshoot > SIP Message Log > Controls**.
The system displays the SIP Message Logging page.
- Step 2** Select if you want to enable or disable SIP message logging.
- Step 3** (Optional) Enter a regular expression filter. This reduces the number of calls that are written to the SIP message log. An example of a regular expression filter is **999...1020**. If you enter this, the system will match any number beginning with 999 and ending with 1020. Only messages that match the regular expression will pass through the filter and be stored.
- Step 4** Click **Update**.

Note In the event of a reload, the log control option in SIP message logging reverts to disabled state and the selected preferences are reset. The user needs to re-assign the preferences.

Searching SIP Message Calls

You can search the SIP message log for certain calls by entering search parameters. If you enter multiple search parameters, the system only returns values that match all the criteria. If you enter no parameters, the system returns all the calls.

There are many SIP messages within each call; if any individual SIP message matches the search criteria, the system displays that call in the search results.



Restriction The system returns a maximum of 500 calls. You can refine the results by entering more search parameters.

SUMMARY STEPS

1. Choose **Troubleshoot** > **SIP Message Log** > **Search Calls**.
2. Enter data on which to search. See [Data for Call Search, on page 191](#).
3. Click **Search**.
4. To clear the SIP message log, click **Clear SIP Message Log**.
5. To see more information about a call that the system returned, click it. The system displays the Call Log page with details about the call.

DETAILED STEPS

-
- Step 1** Choose **Troubleshoot** > **SIP Message Log** > **Search Calls**.
The system displays the SIP Message Log Search page.
- Step 2** Enter data on which to search. See [Data for Call Search, on page 191](#).
- Step 3** Click **Search**.
The system refreshes the page and displays any calls that match the search criteria.
- Step 4** To clear the SIP message log, click **Clear SIP Message Log**.
- Step 5** To see more information about a call that the system returned, click it. The system displays the Call Log page with details about the call.
-

Data for Call Search

Data for Call Search

| Field | Description |
|--|---|
| Called Party—The following parameters apply to the party initiating the call: | |
| Request-URI contains | Matches the supplied string against the SIP Request-URI field in each SIP message |
| Remote Party ID contains | Matches the supplied string against the SIP Remote Party-ID field in each SIP message |
| P-Asserted ID contains | Matches the supplied string against the SIP P-Asserted ID field in each SIP message |
| To header contains | Matches the supplied string against the SIP To Header field in each SIP message |
| Calling Party—The following parameters apply to the party receiving the call: | |
| From: header contains | Matches the supplied string against the SIP From Header field in each SIP message |
| Date and Time—The following parameters limit the search results to an inclusive window of time: | |
| Start Time | Calls before this time are excluded. Note If you enter a value in this field, it must include a time and not just a date. If you do not enter a time, the system returns nothing. |
| End Time | Calls after this time are excluded. Note If you enter a value in this field, it must include a time and not just a date. If you do not enter a time, the system returns nothing. |

Viewing SIP Message Calls

The Call Log page displays the individual SIP messages that were processed by the Cisco Unified SIP Proxy during the dialog. It shows the time the message was handled and the direction relative to the Cisco Unified SIP Proxy.

SUMMARY STEPS

1. Choose **Troubleshoot > SIP Message Log > Search Calls**.
2. Enter data on which to search. See [Searching SIP Message Calls, on page 191](#).
3. Click **Search**.
4. Click on any call.

DETAILED STEPS

- Step 1** Choose **Troubleshoot** > **SIP Message Log** > **Search Calls**.
The system displays the SIP Message Log Search page.
- Step 2** Enter data on which to search. See [Searching SIP Message Calls, on page 191](#).
- Step 3** Click **Search**.
The system refreshes the page and displays any calls that match the search criteria.
- Step 4** Click on any call.
The system displays the Call Log page with details about the call.
-

Related Topics

[Searching SIP Message Calls, on page 191](#)

Enabling the Failed Calls Log

Use the failed calls log to capture and troubleshoot calls that either fail during initial call setup or that do not terminate normally.

The failed calls log is disabled by default. After you enable it, the system automatically generates a log entry for call setup requests that result in a non-successful status. Similarly, calls that do not terminate properly, including calls exceeding the configured session timeout (when call admission control is enabled), will generate a failed calls log entry.



Note You enable the failed calls log independently from the SIP message log. If you want to review the SIP message details for a failed call, enable the SIP message log. See [Enabling SIP Message Logging, on page 190](#).

SUMMARY STEPS

1. Choose **Troubleshoot** > **Failed Calls Log** > **Controls**.
2. Select **Enable** to enable the failed call log.
3. If you want to include calls that failed due to license limitations, check **Log failed calls due to license limit**.
4. Click **Update**.

DETAILED STEPS

- Step 1** Choose **Troubleshoot** > **Failed Calls Log** > **Controls**.
The system displays the Failed Call Logging page.
- Step 2** Select **Enable** to enable the failed call log.
- Step 3** If you want to include calls that failed due to license limitations, check **Log failed calls due to license limit**.

Step 4 Click **Update**.

Related Topics

[Enabling SIP Message Logging](#), on page 190

Viewing the Failed Calls Log

Use the failed calls log to capture and troubleshoot calls that either fail during initial call setup or that do not terminate normally.

SUMMARY STEPS

1. Choose **Troubleshoot** > **Failed Calls Log** > **Search Calls**.
2. To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.
3. To see a different number of failed calls on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, or 100 failed calls.
4. To clear the log, click **Clear All Calls**.

DETAILED STEPS

Step 1 Choose **Troubleshoot** > **Failed Calls Log** > **Search Calls**.

The system displays the Failed Calls Log page and shows the contents of the log file.

Step 2 To move to another page, use the left and right arrow buttons, or enter another page number and press **Enter**.

Step 3 To see a different number of failed calls on each page, on the top right, choose another number from the drop-down box and click **Go**. You can choose to see 10, 25, 50, or 100 failed calls.

Step 4 To clear the log, click **Clear All Calls**.

Viewing the History of a Failed Call

SUMMARY STEPS

1. Choose **Troubleshoot** > **Failed Calls Log**.
2. To see more information about a particular failed call, click the underlined call ID.

DETAILED STEPS

Step 1 Choose **Troubleshoot** > **Failed Calls Log**.

The system displays the Failed Calls Log page and shows the contents of the log file.

Step 2 To see more information about a particular failed call, click the underlined call ID.

The system displays the Failed Call Session History page containing more information about the call.



CHAPTER 35

Error Messages

- [CUSP Internal Error, on page 197](#)
- [Request Not Found, on page 197](#)
- [Authorization Failure, on page 197](#)
- [Configuration Prerequisite Missing, on page 198](#)

CUSP Internal Error

You received this error message because an unexpected internal error has occurred within the Cisco Unified SIP Proxy software.

The web page contains useful details about the problem that occurred. You can provide this information to Cisco TAC.

Try the operation again, and if the problem persists, contact Cisco TAC for assistance.

Request Not Found

You received this error message because the system received an invalid URL page request to the Cisco Unified SIP Proxy web server. If you received this message after clicking a link, it is possible that the Cisco Unified SIP Proxy web server page data is missing or has become corrupt.

If you typed the URL directly into the web browser, double check the exact spelling for typographic errors and try again. If the problem persists, contact Cisco TAC for assistance.

Authorization Failure

You received this error message because you do not have the appropriate privilege to access the web page.

If you believe that you should have permission to access the web page, contact a Cisco Unified SIP Proxy administrator that has superuser privileges. The administrator can modify your user privileges to grant you access to the web page.

Configuration Prerequisite Missing

You received this error message because the system cannot display the web page that you are requesting due to a missing configuration parameter.

The system lists the configuration parameter to be fixed and provides a link to the web page where you can configure the parameter.