

SIP Settings

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Restcomm SMSC comes with default JSLEE SIP RA. You can modify the SIP settings using the CLI or GUI.

Using CLI

You can modify the SIP settings by issuing the command `smc sip modify` with appropriate parameters as described below. You can view the settings by issuing the command `smc sip show` which will display all the values.

Name

`smc sip modify`

SYNOPSIS

```
smc sip modify name cluster-name <clusterName> host <ip> port <port>
routing-ton <routing address ton> routing-npi <routing address npi>
routing-range <routing address range> counters-enabled <true | false>
charging-enabled <true | false> networkid <networkId>
```

DESCRIPTION

This command is used to modify SIP settings.

PARAMETERS

Standard Parameters

- | | |
|--------------|--|
| Name | - The name of the SIP Stack that is being modified. Since the Gateway does not allow creating more than one SIP Stack presently, the name is hardcoded to "SIP". |
| Cluster-name | - The name of the Cluster to which this SIP Stack belongs to. This parameter is not used presently and is meant for future use when Cluster of SIP is enabled. |
| Host | - IP address of the remote node to which all SIP messages must be forwarded. |
| Port | - Port of the remote node to which all SIP messages must be forwarded. |
| routing-ton | - The DefaultSmsRoutingRule will try to match the 'dest_addr_ton' of outgoing SMS with the value configured here. If this configured value is null(-1) or not null and matches, the SMSC will compare the 'dest_addr_npi' and 'destination_addr' as explained below. If it doesn't match, the SMSC will select the next SIP in the list for matching routing rule. |
| routing-npi | - The DefaultSmsRoutingRule will try to match the 'dest_addr_npi' of outgoing SMS with the value |

configured here. If this configured value is null(-1) or not null and matches, the SMSC will compare the 'destination_addr' as below. If it doesn't match, the SMSC will select the next SIP in the list for matching routing rule.

- routing-range - The DefaultSmsRoutingRule will try to match the 'destination_addr' of outgoing SMS with the value configured here. This is a regular java expression and default value is null. If it matches, the SMSC will send the SMS out over this SIP connection. If it doesn't match, the SMSC will select the next ESME in the list for matching routing rule.
- counters-enabled - Flag to enable or disable counters. Not used presently.
- charging-enabled - Flag to enable or disable charging for every SMS arriving from SIP.
- networkId - means to which virtual subnetwork belongs the SIP connector (this is for multi-tenancy support). Default value is 0. If you do not use multi-tenancy support - set this value to 0.

Using GUI

Procedure: Managing the SIP Connection using the GUI

1. In the GUI Management Console for SMSC Gateway, click on 'SIPs' in the left panel.
2. The main panel will display the existing SIP settings. You can view and modify the SIP settings in this panel.