



Oracle Enterprise Session Border Controller –
Acme Packet 4600 and Microsoft Skype for
Business for Enterprise SIP Trunking

Technical Application Note




Disclaimer

The following is intended to outline our general product direction. It is intended for information purposes only, and may not be incorporated into any contract. It is not a commitment to deliver any material, code, or functionality, and should not be relied upon in making purchasing decisions. The development, release, and timing of any features or functionality described for Oracle's products remains at the sole discretion of Oracle.

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Intended Audience

This document is intended for use by Oracle personnel, third party Systems Integrators, and end users of the Oracle Enterprise Session Border Controller (E-SBC). It assumes that the reader is familiar with basic operations of the Oracle Enterprise Session Border Controller – Acme Packet 4600.

Document Overview

Microsoft Skype for Business offers the ability to connect to SIP based telephony trunks using an IP communication. This reduces the cost and complexity of extending an enterprise's telephony system outside its network borders. Oracle Enterprise Session Border Controllers (E-SBCs) play an important role in SIP trunking as they are used by many trunk providers and some enterprises as part of their SIP trunking infrastructure.

This application note has been prepared as a means of ensuring that SIP trunking between Microsoft Skype for Business servers, Oracle E-SBCs and IP Trunking services are configured in the optimal manner.

It should be noted that the E-SBC configuration provided in this guide focuses strictly on the Skype for Business Server associated parameters. Many E-SBC users may have additional configuration requirements that are specific to other applications. These configuration items are not covered in this guide. Please contact your Oracle representative with any questions pertaining to this topic.

For additional information on Skype for Business Server, please visit <http://www.skype.com/en/business/>.

Introduction

Audience

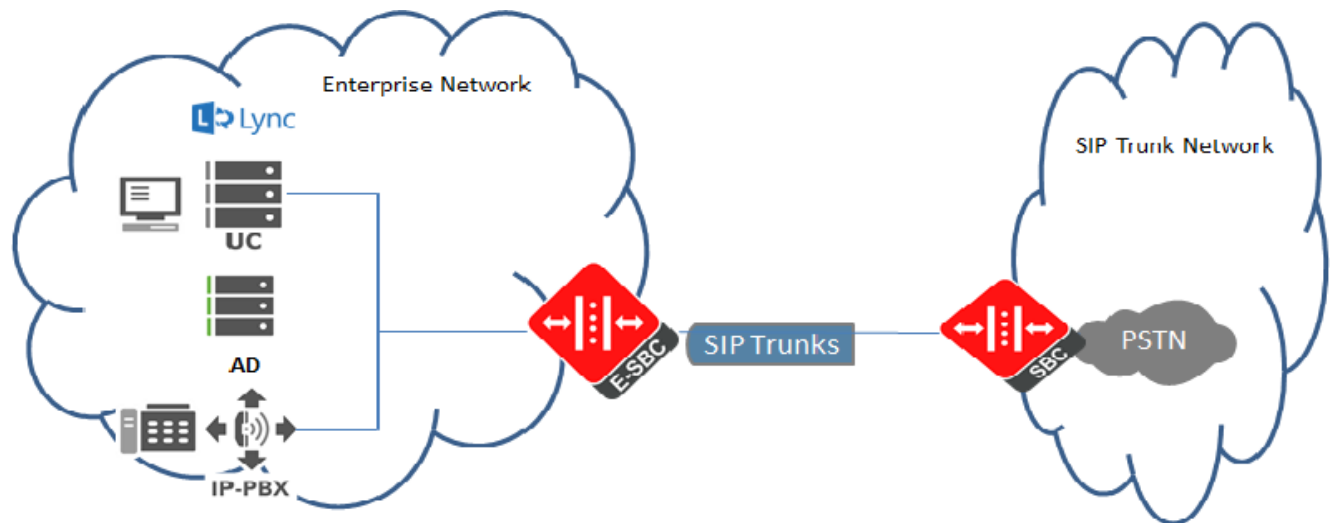
This is a technical document intended for telecommunications engineers with the purpose of configuring the Oracle Enterprise SBC and the Skype for Business Server. There will be steps that require navigating Microsoft Windows Server as well as the Acme Packet Command Line Interface (ACLI). Understanding the basic concepts of TCP/UDP, IP/Routing, and SIP/RTP are also necessary to complete the configuration and for troubleshooting, if necessary.

Requirements

- Fully functioning Skype for Business Server deployment, including Active Directory and DNS
- A dedicated Mediation Server for the SIP trunking connection
- Microsoft Skype for Business 2015 – Version 6.0.93190.0
- Skype for Business 2015 client, Version 15.0.4753.1000
- Oracle Enterprise Session Border Controller AP 4600 running Net-Net OS ECZ730p2.64.bz. Note: the configuration running on the SBC is backward/forward compatible with any release in the 7.3.0 stream.

Architecture

The following reference architecture shows a logical view of the connectivity between Skype for Business Server and the SBC.

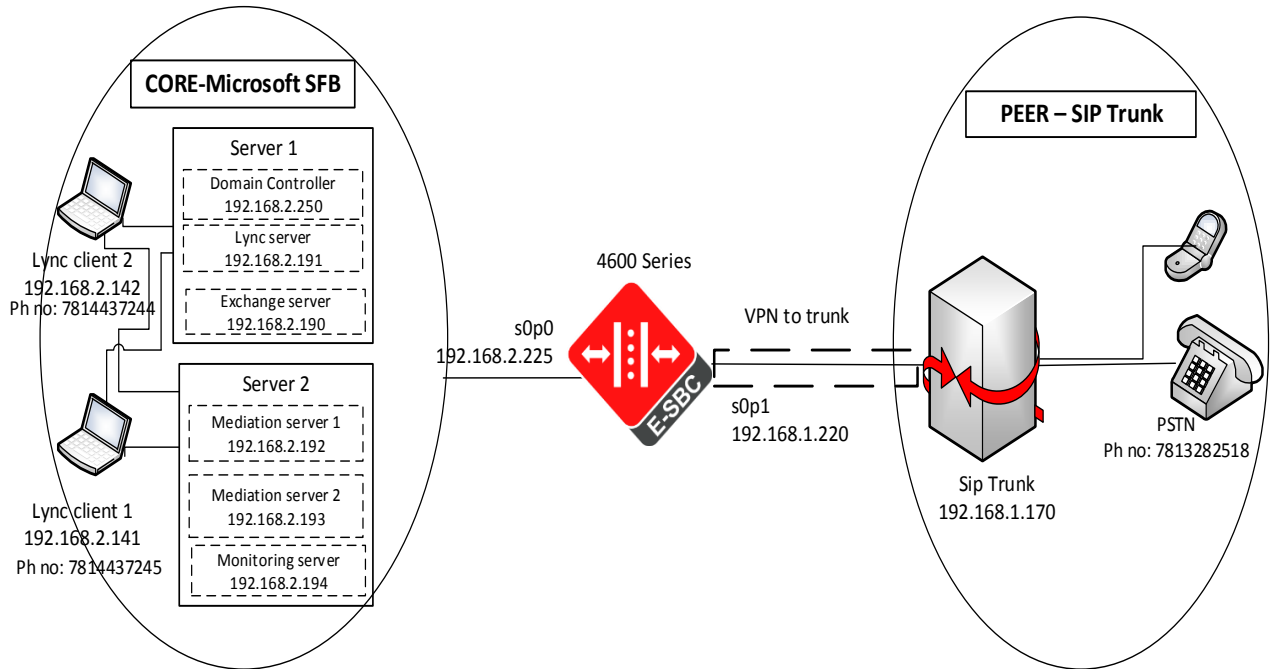


Area A represents the customer's on-premise infrastructure, which includes the Active Directory, DNS and Skype for Business Server systems. Area B represents the service provider infrastructure which provides PSTN service via the SIP trunk. Area C represents the integration of these two environments over an IP network. This could be, through a VPN tunnel over the Internet, an MPLS managed network, or even a dedicated physical connection. The SFB Mediation Server and the SBC are the edge components that form the boundary of the SIP trunk. The configuration, validation and troubleshooting of the areas B and C is the focus of this document and will be described in two phases:

- Phase 1 – Configure the Skype for Business Server
- Phase 2 – Configure the 4600

Lab Configuration

The following diagram, similar to the Reference Architecture described earlier in this document, illustrates the lab environment created to facilitate certification testing (IP addressing/Port below is only a reference, they can change per your network specification).



Phase 1 – Configuring the Skype for Business server

The enterprise will have a fully functioning Lync Server infrastructure with Enterprise Voice deployed and a Mediation Server dedicated to this installation. If there is no Mediation Server present for this purpose, one will have to be deployed.

There are two parts for configuring SFB Server to operate with the Oracle SBC:

- Adding the SBC as a PSTN gateway to the SFB Server infrastructure
- Creating a route within the SFB Server infrastructure to utilize the SIP trunk connected through the SBC.

To add the PSTN gateway, we will need:

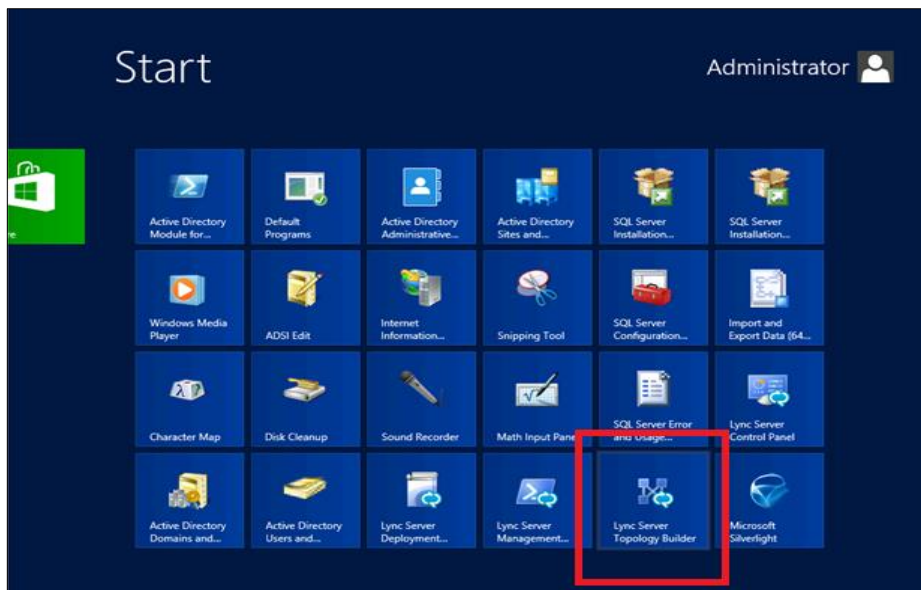
- IP addresses of the external facing NICs of the Mediation Servers
- IP address of the sip interface of the SBC facing the Mediation servers
- Rights to administer Lync Server Topology Builder
- Access to the Lync Server Topology Builder

Note: This section of the Application note only walks you through adding Oracle E-SBC to Microsoft's Skype for Business config, The assumption is that Microsoft's Skype for Business application is already installed and 100% functional.

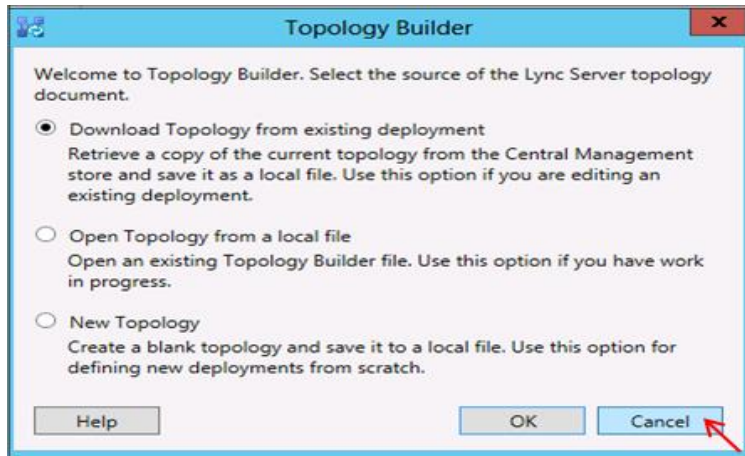
Adding the SBC as a PSTN gateway

The following process details the steps to add the SBC as the PSTN gateway

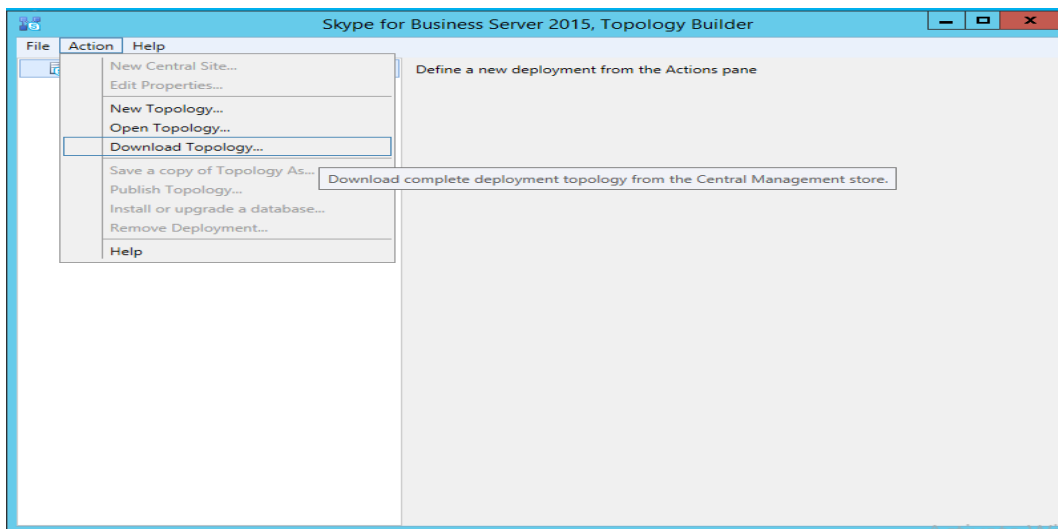
1. On the server where the Topology Builder is located, start the console.
2. From the **Start** bar, select **Lync Server Topology Builder**.



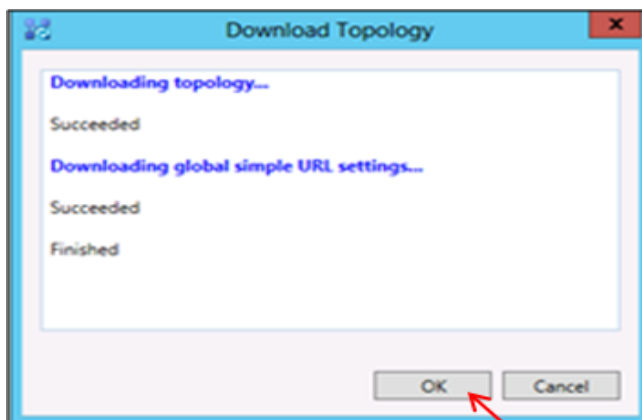
- The opening screen of the Topology builder will be displayed. Click on the **Cancel** button.



- The Topology Builder window will now be displayed. Click on **Action** and select **Download Topology**.

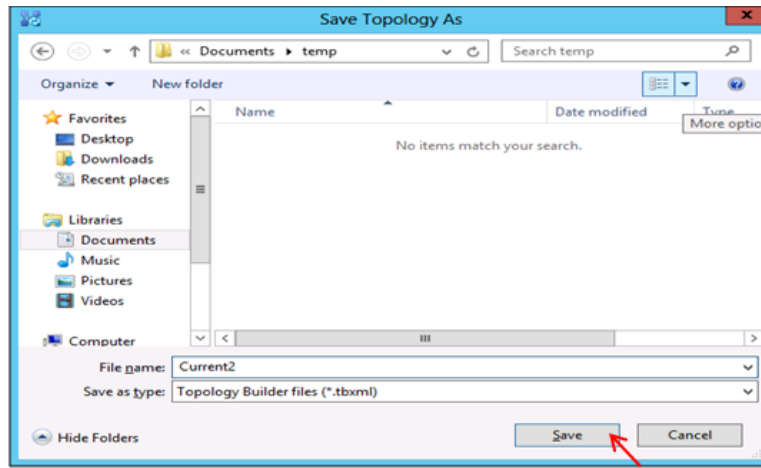


- You will then see a screen showing that you have successfully imported the topology. Click the **Ok** button.

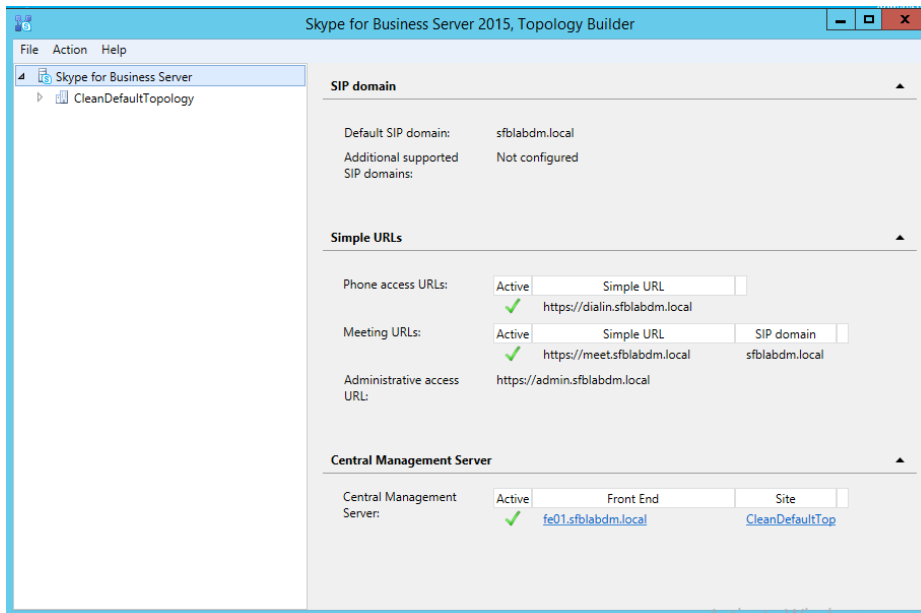


- Next you will be prompted to save the topology which you have imported. You should revision the name or number of the topology according to the standards used within the enterprise. Click the **Save** button

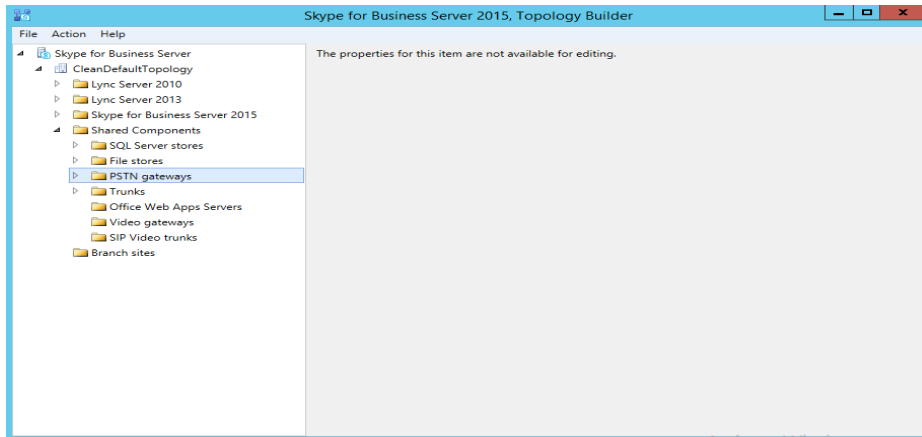
Note: This keeps track of topology changes and, if desired, will allow you to fall back from any changes you make during this installation



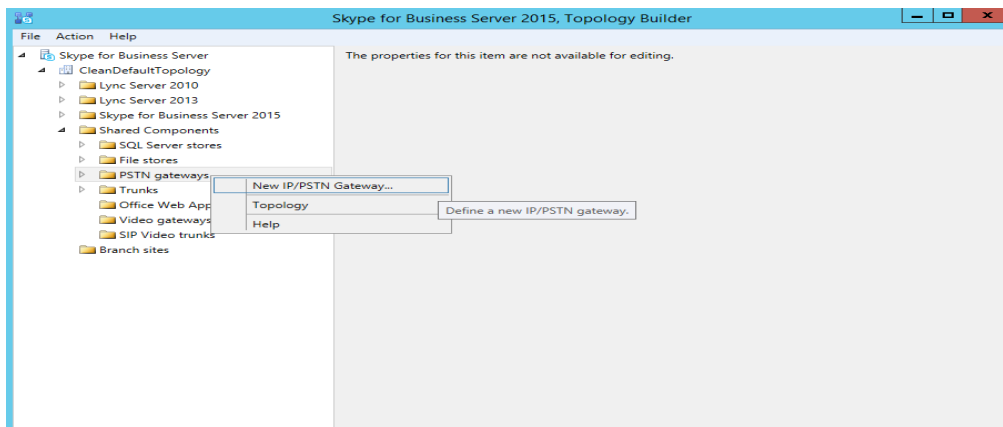
- You will now see the topology builder screen with the enterprise's topology imported.



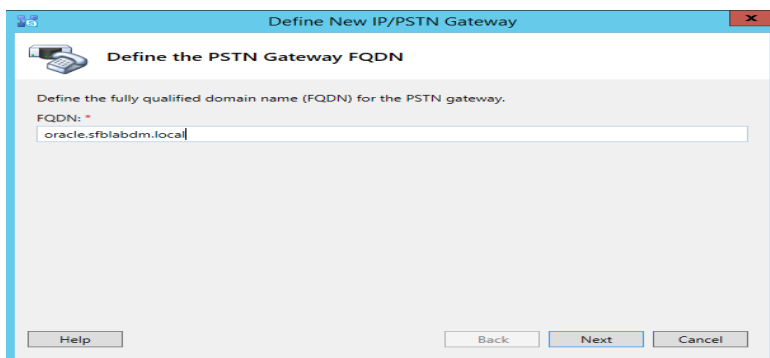
8. In the upper left hand corner, expand the site in which the PSTN gateway will be added. In our case, the site is **Bedford**. Then click on the **PSTN Gateways**



9. Right click on **PSTN gateways** and select **New IP/PSTN Gateway**.



10. In the **Define New IP/PSTN Gateway** window, enter the ip address of the SIP interface of the SBC in the **FQDN** text box and click **Next**.



11. Select **Enable IPv4** in the **Define the IP address** section and click **Next**.

The screenshot shows a dialog box titled "Define New IP/PSTN Gateway" with a close button (X) in the top right corner. The main heading is "Define the IP address" with a telephone icon. There are two radio button options: "Enable IPv4" (selected) and "Enable IPv6". Under "Enable IPv4", there are two sub-options: "Use all configured IP addresses." (selected) and "Limit service usage to selected IP addresses." Below these are two text input fields labeled "PSTN IP address:". At the bottom, there are four buttons: "Help", "Back", "Next" (highlighted in blue), and "Cancel".

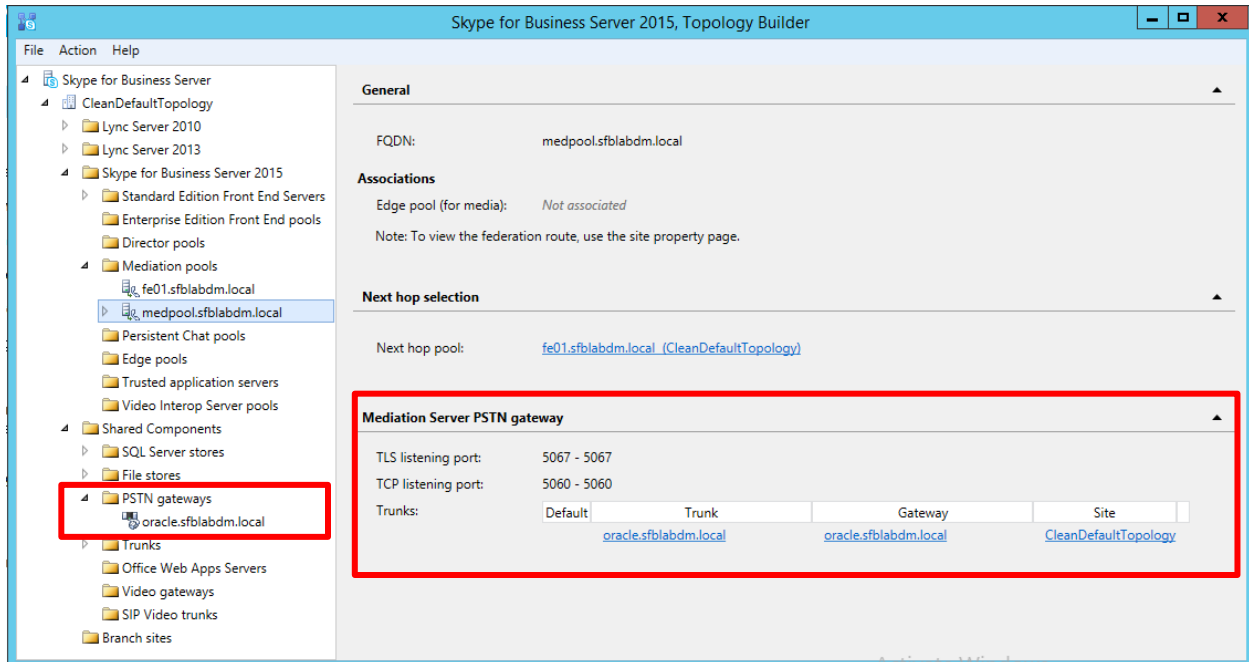
12. In the next section, enter the ip address of the SBC's sip interface under **Trunk name**. Configure the **Listening port for IP/PSTN gateway** as 5068, TCP as the **SIP Transport Protocol** and click **Finish**.

The screenshot shows the same dialog box, now in the "Define the root trunk" section. The "Trunk name:" field contains "oracle.sflabldm.local". The "Listening port for IP/PSTN gateway:" field contains "5068". The "SIP Transport Protocol:" dropdown menu is set to "TCP". The "Associated Mediation Server:" dropdown menu is set to "medpool.sflabldm.local CleanDefaultTopology". The "Associated Mediation Server port:" field contains "5060". At the bottom, there are four buttons: "Help", "Back", "Finish" (highlighted in blue), and "Cancel".

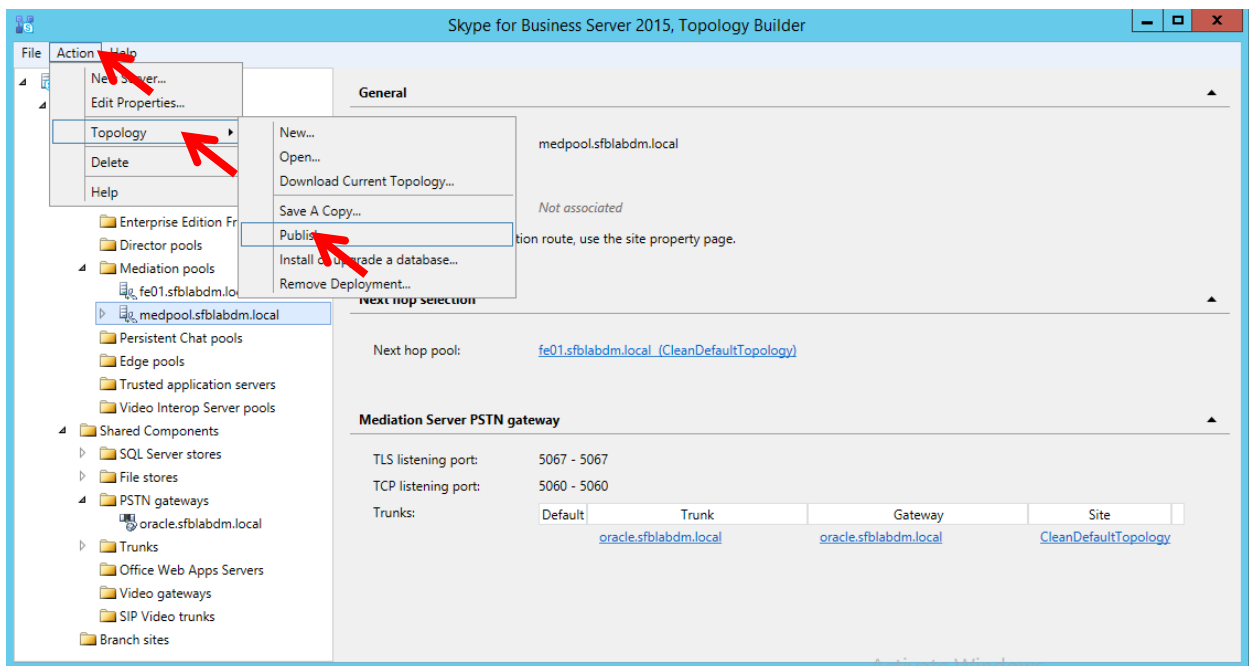
13. The PSTN gateway for the Lync server has been added. It will be listed under **PSTN gateways**.

Expand the **Mediation Pool** list and click on the Mediation Server to be utilized. In our example the Mediation Server is **sfbmedpool.acmepacket.net**.

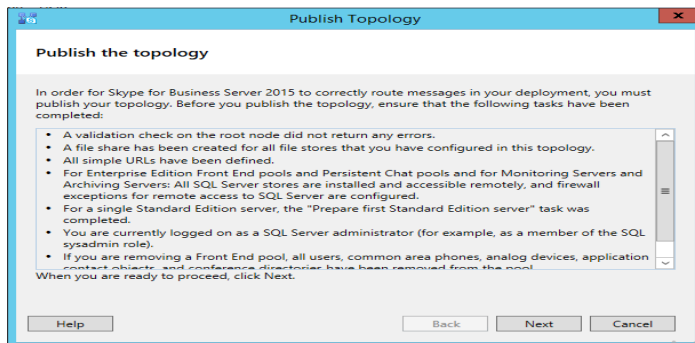
You will see that the PSTN gateway is associated with the Mediation server.



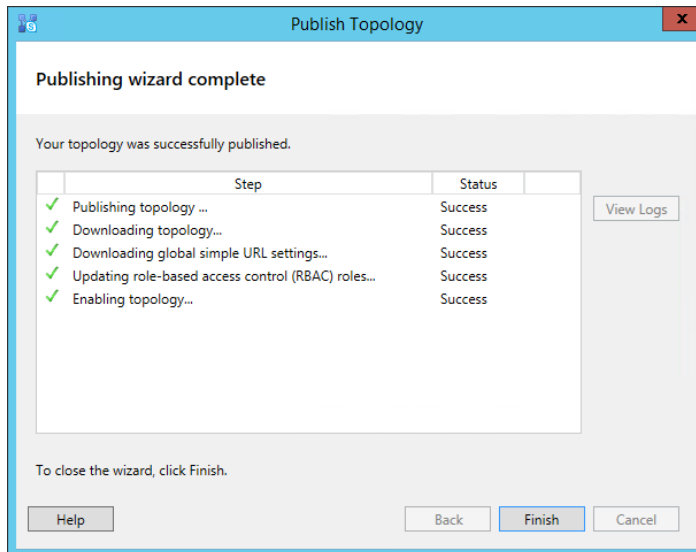
14. In the upper right hand corner of your screen under **Actions** select **Topology** then select **Publish**.



You will now see the **Publish Topology** window. Click on the **Next** button



You will now be at a window showing the databases associated with site. Click **Next**.



15. When complete you should see a window from Topology Builder stating that your topology was successfully published. Click the **Finish** button.

16. You will be at the Topology Builder main window, expand your site and double check that your PSTN entries are correct and that the appropriate Mediation Server has the PSTN gateway associated.

Creating a route within the Lync Server infrastructure

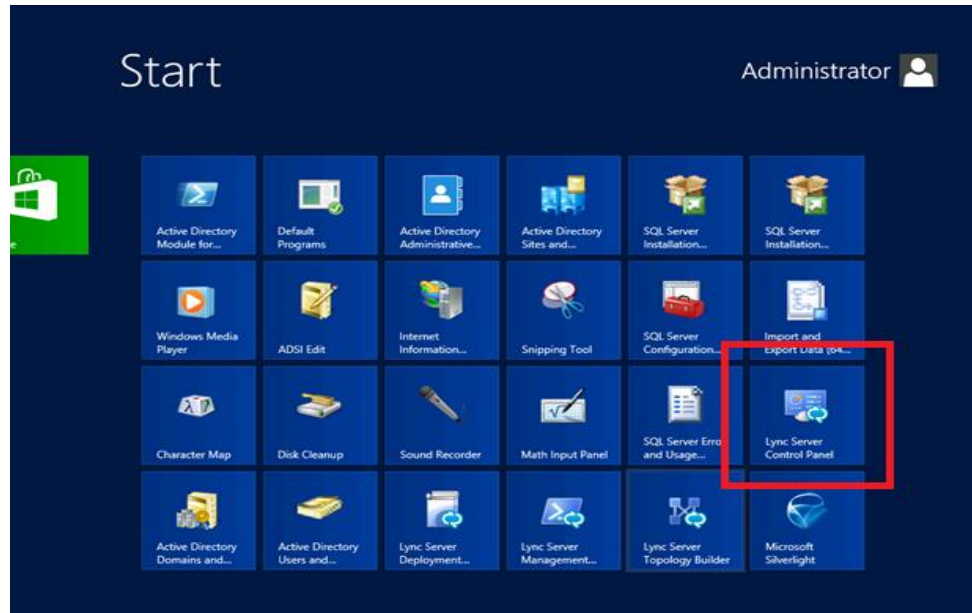
In order for the Lync Server Enterprise Voice clients to utilize the SIP trunking infrastructure that has been put in place, a route will need to be created to allow direction to this egress. Routes specify how Lync Server handles calls placed by enterprise voice users. When a user places a call, the server, if necessary, normalizes the phone number to the E.164 format and then attempts to match that phone number to a SIP Uniform Resource Identifier (URI). If the server is unable to make a match, it applies outgoing call routing logic based on the number. That logic is defined in the form of a separate voice route for each set of target phone numbers listed in the location profile for a locale. For this document we are only describing how to set up a route. Other aspects which apply to Lync Server Enterprise Voice deployments such as dial plans, voice policies, and PSTN usages are not covered.

To add the route we will need:

- Rights to administer Lync Server Control Panel
 - Membership in the CS Administrator Active Directory Group
- Access to the Lync Server Control Panel

The following process details the steps to create the route:

1. From the **Start** bar, select **Lync Server Control Panel**.

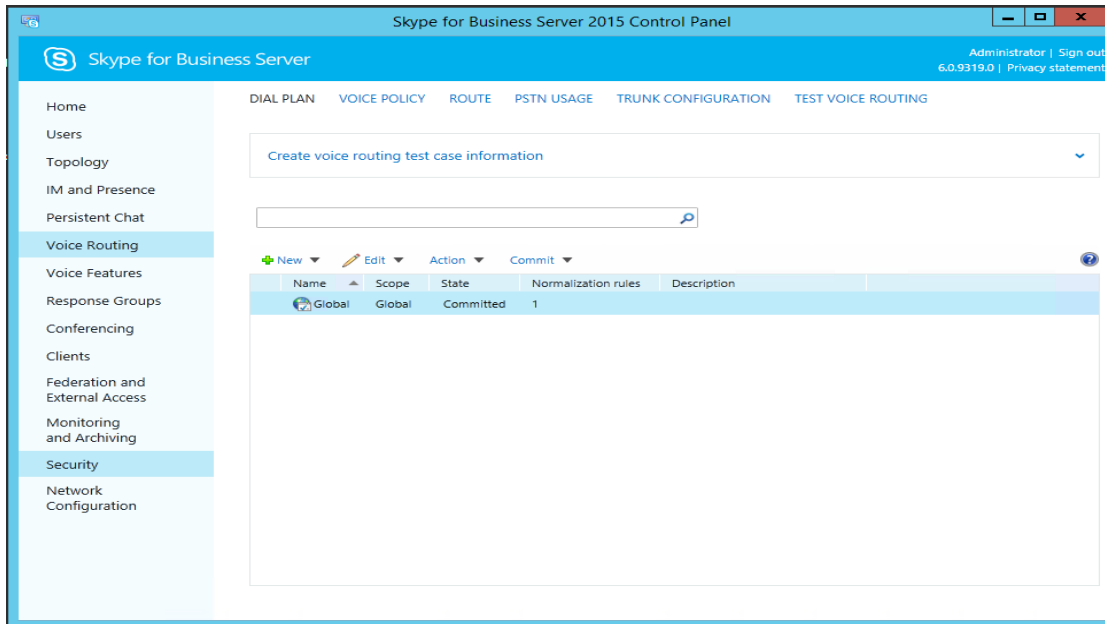


You will be prompted for credential, enter your domain username and password.

2. Once logged in, you will now be at the "Welcome Screen". On the left hand side of the window, click on **Voice Routing**.



3. The **Dial Plan** tab in the **Voice Routing** section will be displayed. On the content area toolbar, click **+New**.

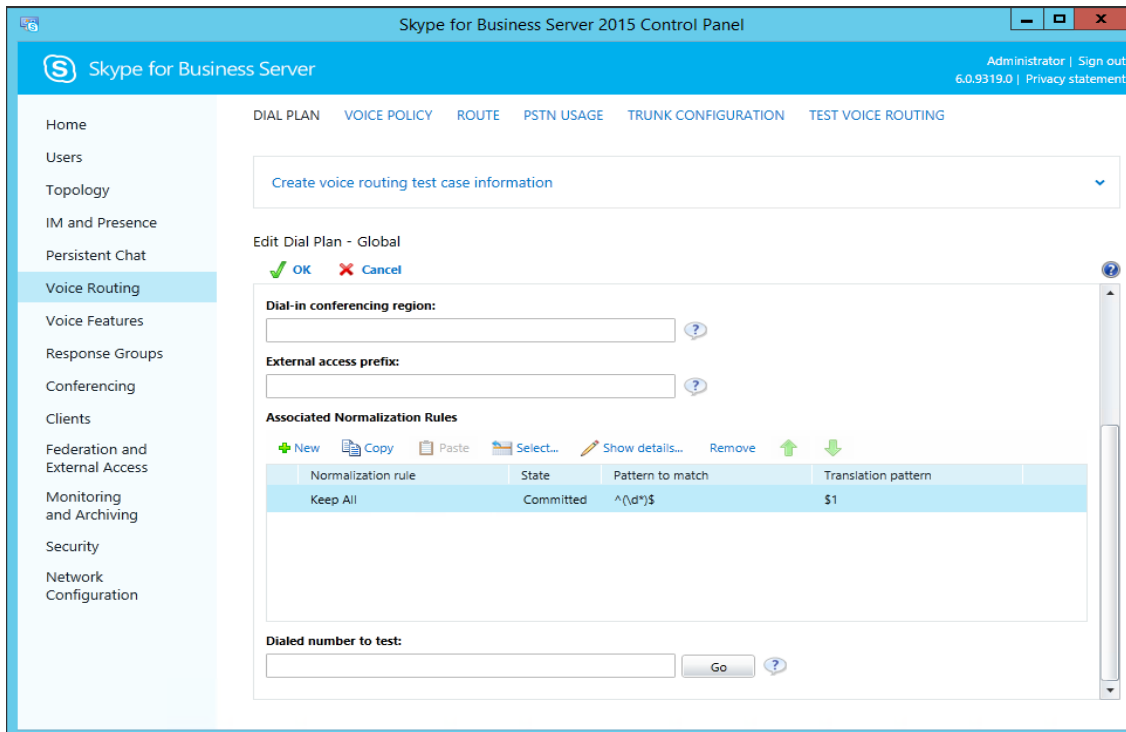


- Next you build a Dial Plan and a translation rule for the phone numbers you want this route to handle. You have to create two separate dial plans for US and EMEA.

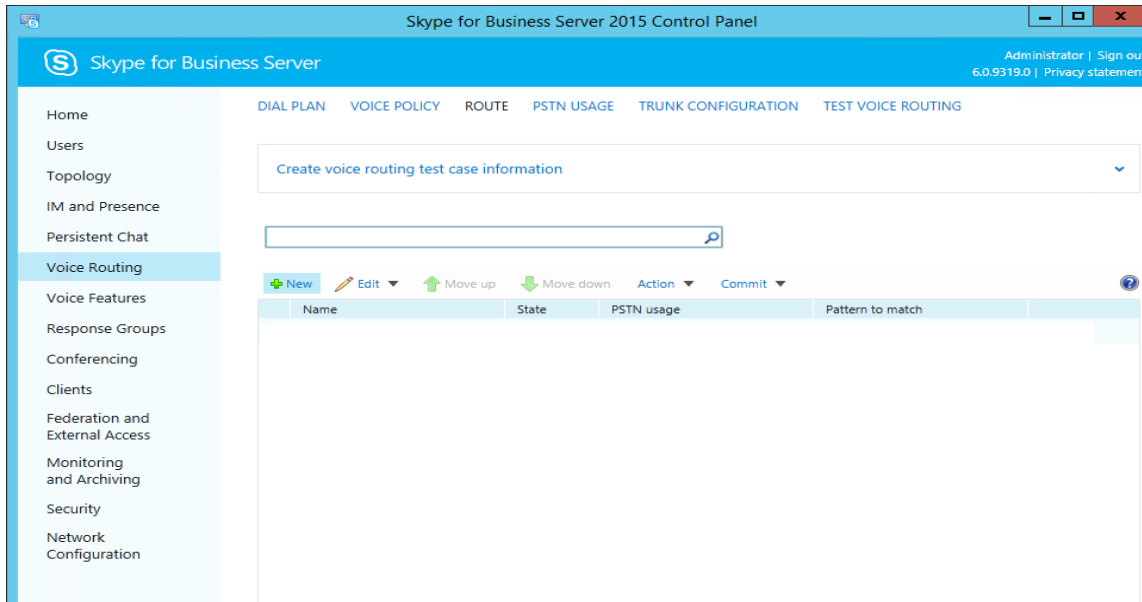
US Dial-plan

Match this pattern: $\wedge(d^*)\$$

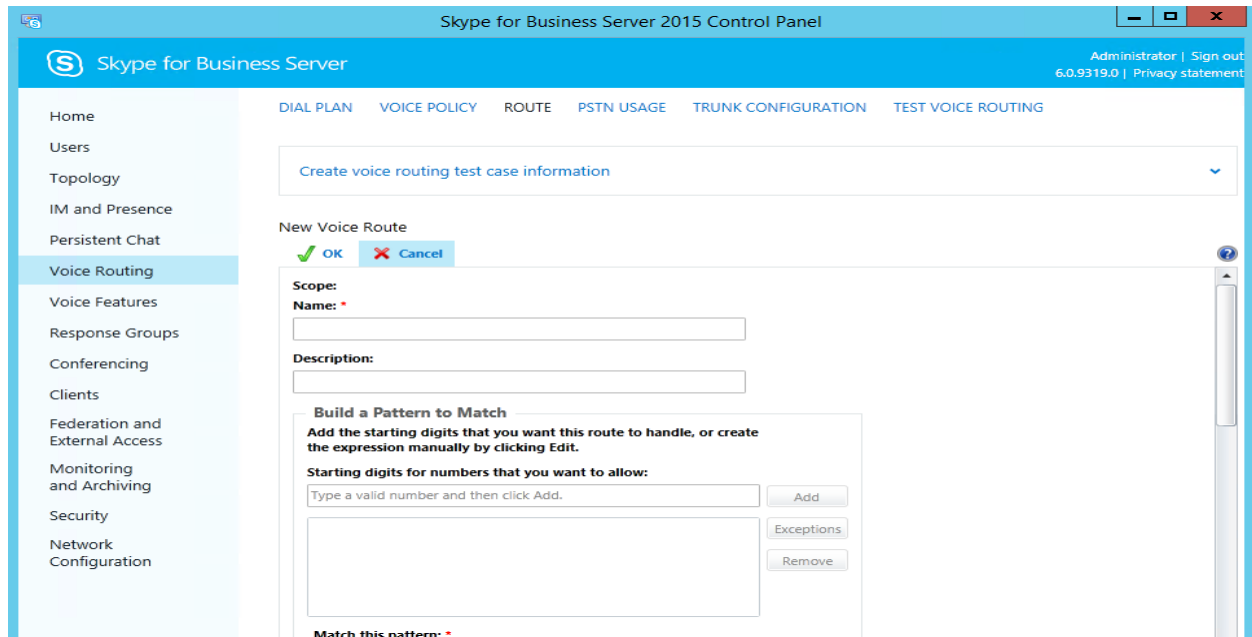
Translation rule: \$1



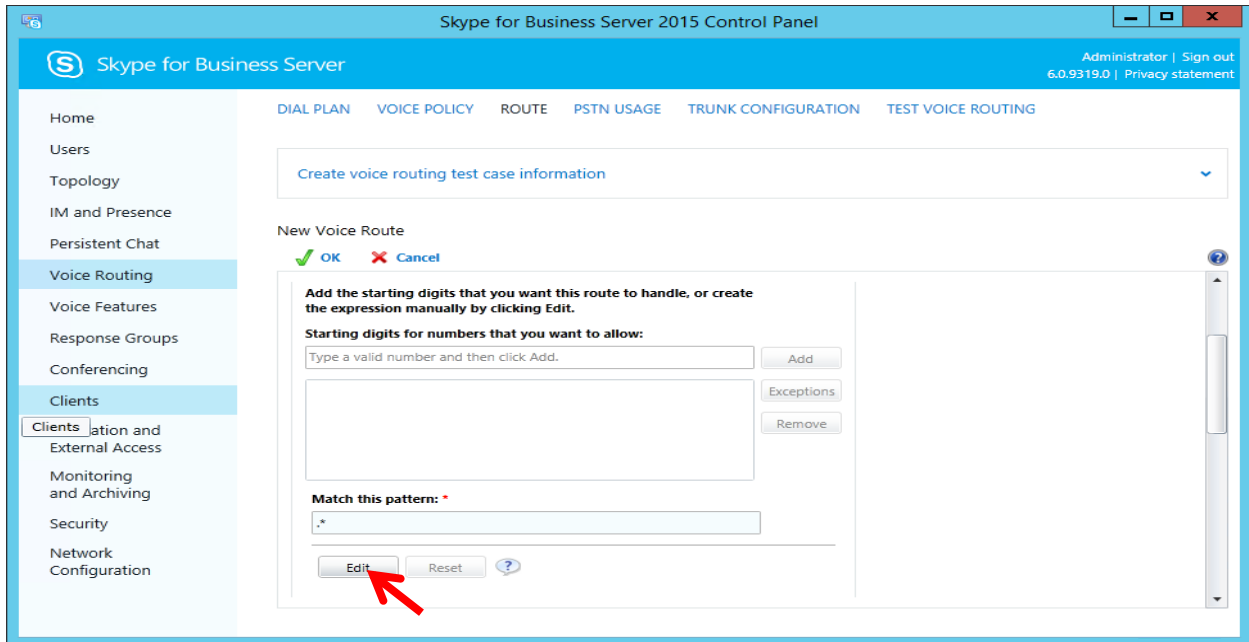
5. On the top row of the tabs, select **Route**. On the content area toolbar, click **+New**.



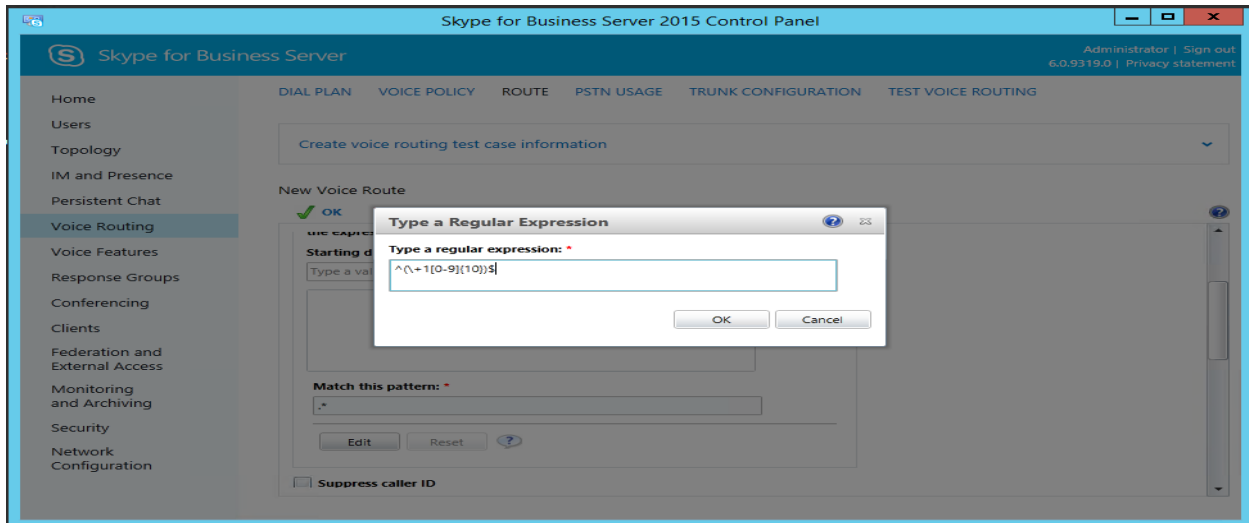
6. On the **New Voice Route** page, in the **Name** field, enter the name you have selected for the Route. In our example, it is **US route**.



7. Next you build a Pattern Match for the phone numbers you want this route to handle. Click **Edit**.



8. Enter the pattern for US - $^{\wedge}\{+1[0-9]{10}\}\$$ and click **OK**.



9. Next you want to associate the Voice Route with the **Trunk** you have just created. Scroll down to **Associated Trunks**, click on the **Add** button.

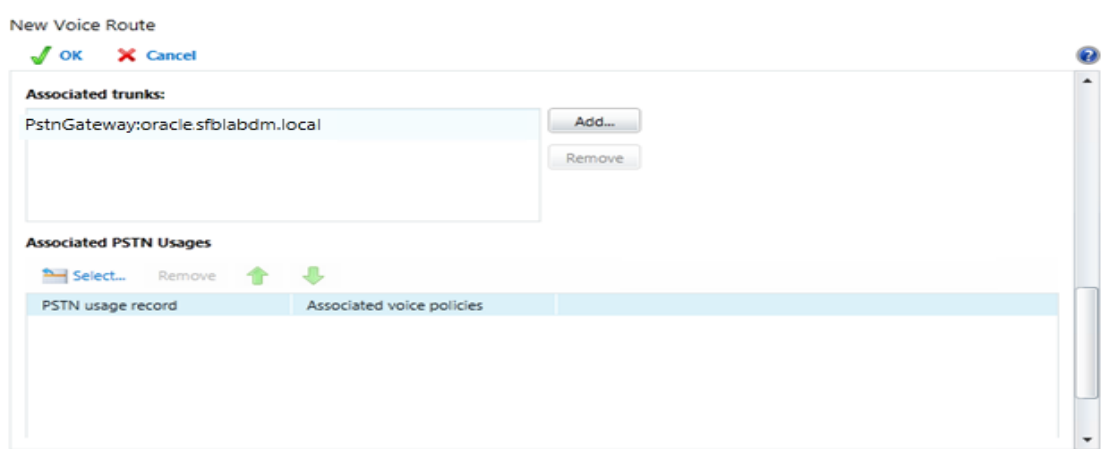
The screenshot shows the 'New Voice Route' configuration page in the Skype for Business Server management console. The page has a blue header with the 'S' logo and 'Skype for Business Server' text. Below the header is a navigation menu with options: DIAL PLAN, VOICE POLICY, ROUTE, PSTN USAGE, TRUNK CONFIGURATION, and TEST VOICE ROUTING. The 'VOICE POLICY' section is active, showing a 'Create voice routing test case information' box. Below this is the 'New Voice Route' section with 'OK' and 'Cancel' buttons. The main configuration area includes a 'Match this pattern:' field with the regex '^(\+1[0-9]{10})\$' and 'Edit', 'Reset', and help buttons. There is a checkbox for 'Suppress caller ID' and an 'Alternate caller ID:' field. At the bottom, the 'Associated trunks:' section has an 'Add...' button and a 'Remove' button.

10. You will now be at a window showing available Trunks to associate your Voice Route. Click on the PSTN gateway that you just created and then click the **OK** button.

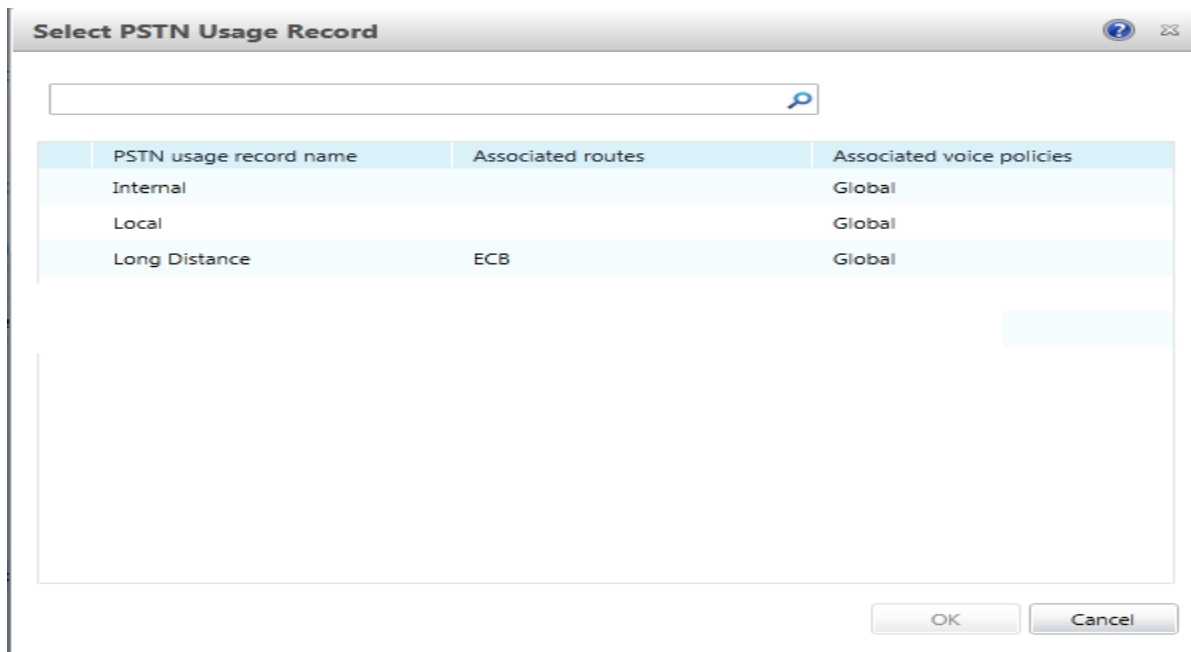
The screenshot shows a 'Select Trunk' dialog box with a search bar at the top. Below the search bar is a table with two columns: 'Service' and 'Site'. The table contains one entry: 'PstnGateway:oracle.sfblabdm.local' under 'Service' and 'CleanDefaultTopology' under 'Site'. At the bottom of the dialog are 'OK' and 'Cancel' buttons.

Service	Site
PstnGateway:oracle.sfblabdm.local	CleanDefaultTopology

11. You can now see that you have associated your trunk with the route you created. An appropriate PSTN usage record will need to be assigned as well. In our example, we use one that was already created in the enterprise. Click on the **Select** button under **Associated PSTN Usages**.



12. In the **Select PSTN Usage Record** window displayed, select the appropriate PSTN Usage Record and click **OK**.



- You will now see the Associated PSTN Usages which you have added. Click the **OK** button at the top of the **New Voice Route** screen.

New Voice Route

✓ OK ✗ Cancel

Associated trunks:

PstnGateway:oracle.sflabldm.local	Add...
	Remove

Associated PSTN Usages

Select... Remove ↑ ↓

PSTN usage record	Associated voice policies
Long Distance	Global

- You will now be at the Routes page showing the US route. Click the **Commit** drop-down menu, and then **Commit All**.

Skype for Business Server Administrator | Sign out 6.0.9319.0 | Privacy statement

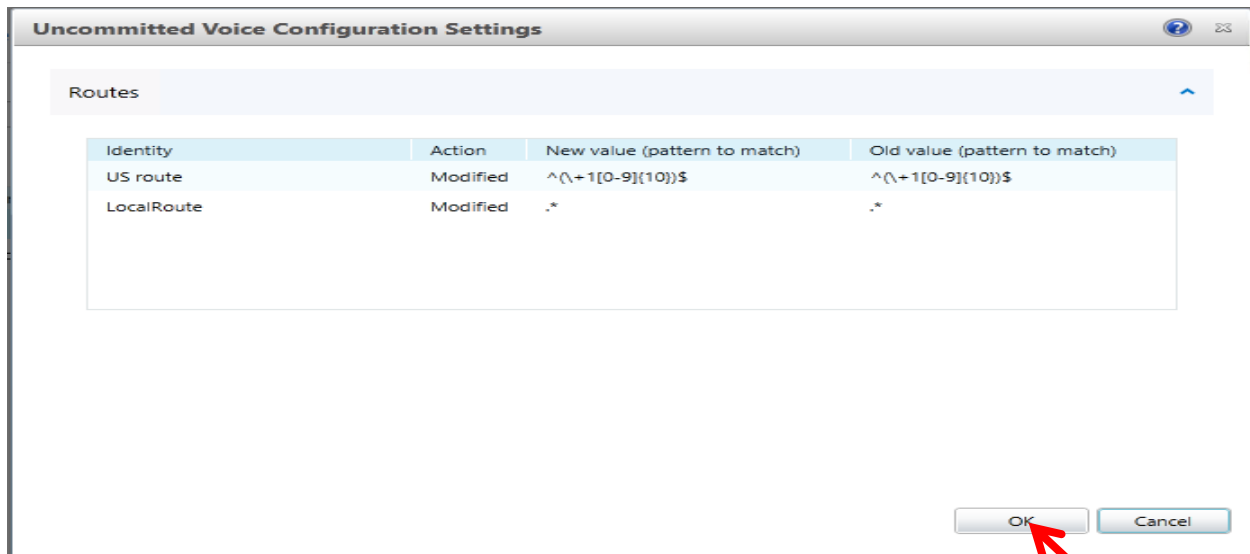
DIAL PLAN VOICE POLICY ROUTE PSTN USAGE TRUNK CONFIGURATION TEST VOICE ROUTING

Create voice routing test case information

Search

Name	State	PSTN usage	Action	Commit
US route	Uncommitted	Long Distar		Review uncommitted changes Commit all
LocalRoute	Uncommitted	PSTN_Usag		Cancel selected changes
ECB	Uncommitted	Long Distar		Cancel all uncommitted changes

15. On the Uncommitted Voice Configuration Settings window, click OK.



If there are no errors, the new Voice Route has now been successfully created and the state will show as **Committed**.

Additional Steps

There are other aspects to a Lync Server Enterprise Voice deployment such as

- Site, local, and global dial plans
- Voice Policies
- Assigning Voice Policies to users
- PSTN usage policies

Refer to [MSDN technet](#) for relevant information.

Phase 2 – Configuring the Oracle Enterprise SBC

In this section we describe the steps for configuring an Oracle Enterprise SBC, formally known as an Acme Packet Net-Net Session Director (“SBC”), for use with Skype for Business Server in a SIP trunking scenario.

In Scope

The following guide configuring the Oracle SBC assumes that this is a newly deployed device dedicated to a single customer. If a service provider currently has the SBC deployed and is adding SFB Server customers, then please see the ACLI Configuration Guide on http://docs.oracle.com/cd/E61547_01/index.html for a better understanding of the Command Line Interface (CLI).

Note that Oracle offers several models of SBC. This document covers the setup for the 4600 platform series running Net-Net OS ECZ7.3.0 or later. If instructions are needed for other Oracle SBC models, please contact your Oracle representative.

Out of Scope

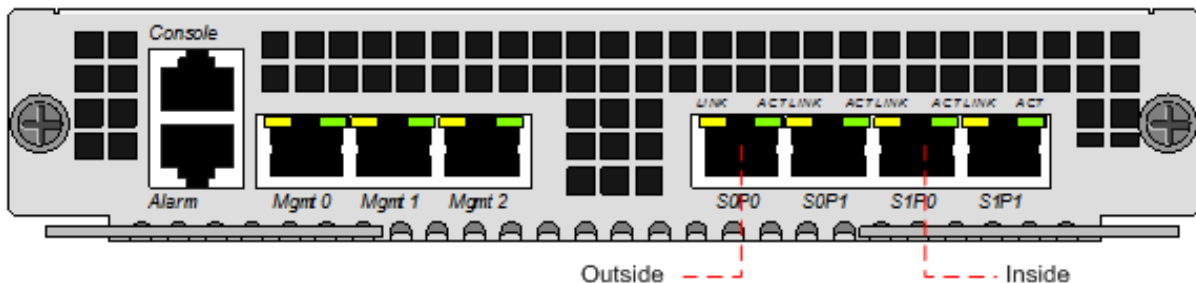
- Configuration of Network management including SNMP and RADIUS; and

What will you need

- Serial Console cross over cable with RJ-45 connector
- Terminal emulation application such as PuTTY or HyperTerm
- Passwords for the User and Superuser modes on the Oracle SBC
- IP address to be assigned to management interface (Wancom0) of the SBC - the Wancom0 management interface must be connected and configured to a management network separate from the service interfaces. Otherwise the SBC is subject to ARP overlap issues, loss of system access when the network is down, and compromising DDoS protection. Oracle does not support SBC configurations with management and media/service interfaces on the same subnet.
- IP address of Mediation Server external facing NIC
- IP addresses to be used for the SBC internal and external facing ports (Service Interfaces)
- IP address of the next hop gateway in the service provider network
- IP address of the enterprise DNS server

SBC- Getting Started

Once the Oracle SBC is racked and the power cable connected, you are ready to set up physical network connectivity. **Note: use the console port on the front of the SBC, not the one on the back.**



Plug the slot 0 port 0 (s0p0) interface into your outside (gateway facing) network and the slot 0 port 1 (s1p0) interface into your inside (SFB server-facing) network. Once connected, perform you are ready to power on and perform the following steps.

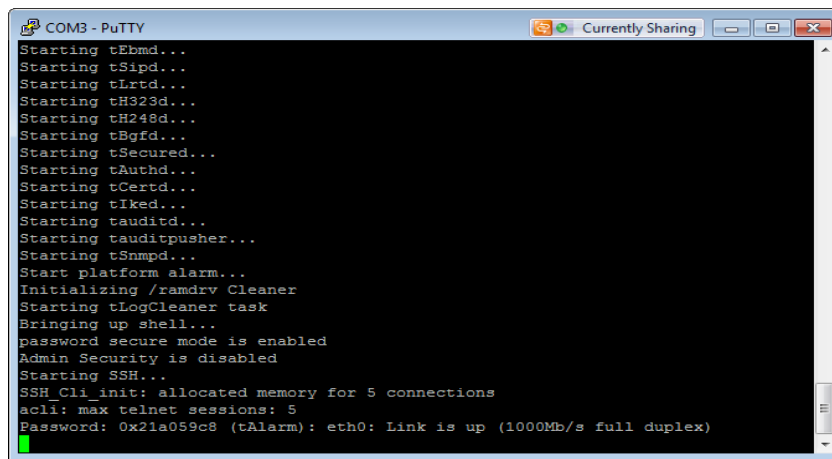
All commands are in bold, such as **configure terminal**; parameters in bold red such as **oraclesbc1** are parameters which are specific to an individual deployment. **Note:** The ACLI is case sensitive.

Establish the serial connection and logging in the SBC

Confirm the SBC is powered off and connect one end of a straight-through Ethernet cable to the front console port (which is active by default) on the SBC and the other end to console adapter that ships with the SBC, connect the console adapter (a DB-9 adapter) to the DB-9 port on a workstation, running a terminal emulator application such as PuTTY. Start the terminal emulation application using the following settings:

- Baud Rate=115200
- Data Bits=8
- Parity=None
- Stop Bits=1
- Flow Control=None

Power on the SBC and confirm that you see the following output from the bootup sequence.



```
COM3 - PuTTY
Starting tEbmnd...
Starting tSipd...
Starting tLrtd...
Starting tH323d...
Starting tH248d...
Starting tBgfd...
Starting tSecured...
Starting tAuthd...
Starting tCerd...
Starting tIked...
Starting tauditd...
Starting tauditpusher...
Starting tSnmpd...
Start platform alarm...
Initializing /ramdrv Cleaner
Starting tLogCleaner task
Bringing up shell...
password secure mode is enabled
Admin Security is disabled
Starting SSH...
SSH_Cli_init: allocated memory for 5 connections
acl: max telnet sessions: 5
Password: 0x21a059c8 (tAlarm): eth0: Link is up (1000Mb/s full duplex)
```

Enter the following commands to login to the SBC and move to the configuration mode. Note that the default SBC password is “acme” and the default super user password is “packet”.

```
Password: acme
oraclesbc1> enable
Password: packet
oraclesbc1# configure terminal
oraclesbc1(configure)#
```

You are now in the global configuration mode.

Initial Configuration – Assigning the management Interface an IP address

To assign an IP address, one has to configure the bootparams on the SBC by going to

oraclesbc1#configure terminal --- >bootparams

- Once you type “bootparam” you have to use “carriage return” key to navigate down
- A reboot is required if changes are made to the existing bootparams

```
ACMESYSTEM(configure)# bootparam

'.' = clear field; '-' = go to previous field; q
= quit
```



```
Boot File           : /boot/nxECZ730p2.64.bz
IP Address          : 192.65.79.44
VLAN                :
Netmask             : 255.255.255.224
Gateway            : 192.65.79.33
IPv6 Address        :
IPv6 Gateway        :
Host IP             : 0.0.0.0
FTP username        : vxftp
FTP password        : vxftp123
Flags               :
Target Name         : ACMESYSTEM
Console Device      : COM1
Console Baudrate    : 115200
Other               :
```

NOTE: These changed parameters will not go into effect until reboot.
Also, be aware that some boot parameters may also be changed through
PHY and Network Interface Configurations.

Configuring the SBC

The following section walks you through configuring the Oracle Enterprise SBC required to work with Skype for Business (SFB) and a SIP trunk. The information below is split into three sections:

- The transport protocol for signaling/media between SBC and SFB is TLS and SRTP – the SIP trunk used doesn't support these transport mechanisms, the SBC interworks TLS & SRTP to UDP & RTP.
- The transport protocol for signaling/media between SBC and SFB is TCP and RTP – the SIP trunk used doesn't support these transport mechanisms, the SBC interworks TCP to UDP for signaling traffic.
- A third section depicts only the IPv6 configuration which includes two sections:
 - TLS with SFB and IPv4 using UDP transport towards the trunk. *Note this configuration requires the "IPv4-v6 Interworking" license.*
 - TCP with SFB and IPv4 using UDP transport towards the trunk. *Note this configuration requires the "IPv4-v6 Interworking" license.*

It is outside the scope of this document to include all the interoperability working information as it will differ in every deployment.

High Availability

The Mgmt1 and Mgmt2 (labeled wancom1 and wancom2 in the configuration) ports which are on the rear panel of the SBC are used for the purpose of High Availability on the 4600s. Crossover cables must be connected between these ports on the SBCs, i.e. Mgmt1 to Mgmt1 and Mgmt2 to Mgmt2. Please refer to the "High Availability Nodes" in the ACLI configuration guide for ECZ730 for more details.

We have also configured some options and sip-manipulations which are specific to any SFB deployment with the Oracle Communications SBC. They are explained below

SIP PRACK interworking

In order to establish an early media session for outbound calls, Skype for Business gateway specification mandates the PSTN gateways to offer a reliable provisional response and for inbound calls offer INVITEs with a supported header. The SBC interworks

the messaging and provide RFC 3262 PRACK interworking towards SFB and it is a mandatory configuration in all Oracle ESBC–Microsoft SFB deployments. The following need to be configured on ESBC

- Configure option 100rel-interworking on the sip-interface facing mediation server
- Configure a sip-feature to pass the 100rel in supported and require headers
- Configure a sip-manipulation (ForEarlyMedia) to add a Require:100rel header in incoming SIP INVITE from mediation server and delete the Supported:100rel header

SIP manipulations

Skype for Business typically sends mediation server FQDN in the Contact header with no username in the SIP URI which when the SBC forwards, is not acceptable by SIP trunk providers. The SBC therefore rewrites the Contact header to include the username appropriately. A sip-manipulation, ChangeContact, will need to be configured to change the format of the CONTACT header. Another sip-manipulation NATting is configured which is referenced in the NATandChangeContact rule which is mainly for NATting the To, From, and Request-URI headers and is applied on the out-manipulationid for both the sip-interfaces. Other sip-manipulations are also outlined in the configuration below.

SFB and SIP trunk carry their own SIP and SDP design – not always these implementation methods align, causing a lot of mismatch in SIP and SDP signaling and call flow. The ESBC helps resolve these issues with SIP manipulation feature. Here are a list of manipulations that were created to make SFB complaint to the SIP Trunk and vice versa.

1. Addpcmato183 --- Adds PCMA to a 183 response
2. ChangeContact --- Fixes the contact header offered by SFB before the message is sent to Trunk
3. Changeinactosendonly ---- SBC changes SDP from inactive to sendonly on INVITEs for hold (required to trigger audio playback from SIP Trunk)
4. Check183 --- Check the response to INVITE is 183 session progress
5. Modcline --- Check if the SDP contains IP of the mediation server – if it does, don't change 183 to 180
6. NATting --- NAT From & To header with correct IP information
7. OptionsResponseLocally --- Respond to OPTIONS locally on the SBC.
8. addPAIOnCallForwards --- Add P-Asserted-Identity header to INVITE on call transfers
9. convert183to180 --- Convert 183 to 180 for triggering early media
10. fix183sdp --- If PCMA is offered, fix the SDP with PCMA + PCMU
11. modcontact --- Modify contact for anonymous calls
12. outManipToSFB --- a few manipulations before message is delivered to SFB
13. replacemtimewithptime --- change maxptime to ptime
14. ForEarlyMedia --- To locally handle PRACK interworking
15. NATandChangeContact --- Manipulate few headers with correct IP information
16. addPAIRefers --- Add P-Asserted-Identity header to INVITE if the REFER message contains Referred-by header

Local REFER handling by the SBC

In event of transferring calls from a SFB client (calls could be initially placed between 2 SFB Clients, but then transferred out to PSTN), SFB has two OPTIONS:

- Transfer using re-INVITE approach
- Transfer using REFER method --- the info below is required when REFER based transfer is enabled.

Oracle E-SBC supports REFER method termination when received from SFB. After termination the ESBC uses the info in refer-to header and sends a newly generated INVITE towards the SFB Mediation server. SFB then processes the INVITE, authorizes the call transfer and sends a new INVITE (for calls transferred to PSTN) to the SBC or transfers call internally to the transferred SFB client

To handle the call transfer and refer scenarios – when SFB client 1 refers/transfers the call to SFB Client 2 or to a party on the PSTN, we will need two routes to route to the two mediation servers depending on the referred party:

```

local-policy
  from-address          *
  to-address            sfbmedpool.acmepacket.net
  source-realm          SIP-Trunk
  description           For referred party header
  activate-time
  deactivate-time
  state                 enabled
  policy-priority       none
  policy-attribute
    next-hop            sfbmedpool.acmepacket.net
    realm               towards-sfb
    action               replace-uri
    terminate-recursion disabled
    carrier
    start-time          0000
    end-time             2400
    days-of-week        U-S
    cost                 0
    state                enabled
    app-protocol        SIP
    methods
    media-profiles
    lookup               single
    next-key
    eloc-str-lkup        disabled
    eloc-str-match

```

Ring-back tone during Transfers

During call transfer to a PSTN party, the transfer completes but the calling party does not hear a ring back tone during transfer. The INVITE Lync sends to the SBC to initiate the transfer contains the SDP attribute, a=inactive which is forwarded to the trunk and as a result of which the SBC cannot play the ring back tone to the original PSTN caller (while call is being transferred). A sendonly attribute is required for MoH and transfer scenarios for the calling party to be able to hear ringback or MoH when it is kept on hold. The SBC is able to signal appropriately towards the SIP trunk by changing the "a=inactive" SDP attribute in the INVITE to sendonly towards PSTN.

Sip manipulations are configured to make the necessary changes. The manipulation ForEarlyMedia is configured to change the SDP attribute from a=inactive to a=sendonly in the INVITEs sent to the calling party for transfer

```

sip-manipulation
  name                  Changeinactosendonly
  description           Change inactive to sendonly for transfer
  split-headers
  join-headers
  header-rule
    name                changeSDP
    header-name          Content-Type
    action               manipulate
    comparison-type      case-sensitive
    msg-type             request
    methods              INVITE
    match-value
    new-value
    element-rule
      name               inacttosendonly
      parameter-name     application/sdp

```

type	mime
action	find-replace-all
match-val-type	any
comparison-type	pattern-rule
match-value	a=inactive
new-value	a=sendonly

We utilize the local playback feature of the SBC to play ring back tone during transfers. The ringback tone is played based on REFER termination. You must upload an audio file (in .raw format) to /code/media onto the ESBC for the media you want played during the transfer. A separate file is required for each different codec type, even if the media itself is the same.

The playback configuration is defined listing the media files that you want to play. The playback-config element is configured under media-manager.

```

playback-config
  name                transferrbt
  entry
    encoding          PCMU
    filename           US_ringbackPCMU.raw
    bytes-per-sec     8000

```

The playback options can be applied to realms, sip-interfaces or session agents using the `spl-options` command.

```

oraclesbcl(session-router)# sip-interface
oraclesbcl(sip-interface)# sel
<realm-id>:
1: towards-sfb 192.168.2.225:5060
2: SIP-trunk 192.168.1.220:5060

selection: 1
oraclesbcl(sip-interface)# spl-options playback-on-refer="transferrbt"
oraclesbcl(sip-interface)# done

```

SBC Configurations

IPv4 based signaling for communication between Skype for Business and Oracle ESBC

IPv4 with TLS/SRTP

Following is the complete TLS/SRTP configuration of the SBC:

```

certificate-record
  name                CAcert
  country             US
  state              MA
  locality           Burlington
  organization       Tekvizion
  unit
  common-name        sfblabdm-DC-CA-1
  key-size           2048
  alternate-name
  trusted            enabled
  key-usage-list     digitalSignature
                   keyEncipherment
  extended-key-usage-list serverAuth
  options
  last-modified-by   admin@IP
  last-modified-date 2015-09-18 15:19:22
certificate-record

```

```

name SBCcert1
country US
state MA
locality Burlington
organization Tekvizion
unit
common-name oracle.sfblabdm.local
key-size 2048
alternate-name oraclepool.sfblabdm.local
trusted enabled
key-usage-list digitalSignature
keyEncipherment
extended-key-usage-list serverAuth
options
last-modified-by admin@IP
last-modified-date 2015-09-18 15:06:39
codec-policy
name For488
allow-codecs G729 PCMU:no telephone-event PCMA:no
add-codecs-on-egress
order-codecs G729 telephone-event
packetization-time 20
force-ptime disabled
dtmf-in-audio disabled
last-modified-by admin@IP
last-modified-date 2015-09-16 17:15:37
codec-policy
name TC
allow-codecs PCMA telephone-event PCMU:no
add-codecs-on-egress PCMA telephone-event G729:no
order-codecs PCMA telephone-event G729:no
packetization-time 20
force-ptime disabled
dtmf-in-audio disabled
last-modified-by admin@IP
last-modified-date 2015-09-11 14:32:23
codec-policy
name TCPCMU
allow-codecs PCMU telephone-event PCMA:no
add-codecs-on-egress
order-codecs PCMU telephone-event
packetization-time 20
force-ptime disabled
dtmf-in-audio disabled
last-modified-by admin@IP
last-modified-date 2015-09-11 14:25:11
local-policy
from-address *
to-address *
source-realm SIP-Trunk
description
activate-time
deactivate-time
state enabled
policy-priority none
policy-attribute

```

```

        next-hop                Medpool.sfbblabdm.local
        realm                    towards-sfb
        action                    replace-uri
        terminate-recursion      disabled
        carrier
        start-time                0000
        end-time                  2400
        days-of-week              U-S
        cost                      0
        state                    enabled
        app-protocol              SIP
        methods
        media-profiles
        lookup                    single
        next-key
        eloc-str-lkup             disabled
        eloc-str-match
    last-modified-by             web_admin@IP
    last-modified-date           2015-09-21 13:23:44
local-policy
    from-address                 *
    to-address                   med1.sfbblabdm.local
    source-realm                 SIP-Trunk
    description                  For Referred-Party Header
    activate-time
    deactivate-time
    state                        enabled
    policy-priority              none
    policy-attribute
        next-hop                med1.sfbblabdm.local
        realm                    towards-sfb
        action                    replace-uri
        terminate-recursion      disabled
        carrier
        start-time                0000
        end-time                  2400
        days-of-week              U-S
        cost                      0
        state                    enabled
        app-protocol              SIP
        methods
        media-profiles
        lookup                    single
        next-key
        eloc-str-lkup             disabled
        eloc-str-match
    last-modified-by             admin@IP
    last-modified-date           2015-09-01 12:31:09
local-policy
    from-address                 *
    to-address                   med2.sfbblabdm.local
    source-realm                 SIP-Trunk
    description
    activate-time
    deactivate-time
    state                        enabled

```

```

policy-priority none
policy-attribute
  next-hop med2.sfbblabdm.local
  realm towards-sfb
  action replace-uri
  terminate-recursion disabled
  carrier
  start-time 0000
  end-time 2400
  days-of-week U-S
  cost 0
  state enabled
  app-protocol
  methods
  media-profiles
  lookup single
  next-key
  eloc-str-lkup disabled
  eloc-str-match
last-modified-by web_admin@IP
last-modified-date 2015-09-01 13:01:08
local-policy
  from-address *
  to-address *
  source-realm towards-sfb
  description
  activate-time
  deactivate-time
  state enabled
  policy-priority none
  policy-attribute
    next-hop 63.87.147.48
    realm SIP-Trunk
    action replace-uri
    terminate-recursion disabled
    carrier
    start-time 0000
    end-time 2400
    days-of-week U-S
    cost 0
    state enabled
    app-protocol SIP
    methods
    media-profiles
    lookup single
    next-key
    eloc-str-lkup disabled
    eloc-str-match
last-modified-by web_admin@IP
last-modified-date 2015-09-01 13:01:21
media-manager
  state enabled
  latching enabled
  flow-time-limit 86400
  initial-guard-timer 300
  subsq-guard-timer 300

```

```

tcp-flow-time-limit          86400
tcp-initial-guard-timer      300
tcp-subsq-guard-timer        300
tcp-number-of-ports-per-flow 2
hnt-rtcp                     disabled
algd-log-level               NOTICE
mbcd-log-level               NOTICE
options
red-flow-port                1985
red-mgcp-port                1986
red-max-trans                10000
red-sync-start-time          5000
red-sync-comp-time           1000
media-policing               enabled
max-signaling-bandwidth      10000000
max-untrusted-signaling      100
min-untrusted-signaling      30
tolerance-window             30
trap-on-demote-to-deny       disabled
trap-on-demote-to-untrusted   disabled
syslog-on-demote-to-deny     disabled
syslog-on-demote-to-untrusted disabled
rtcp-rate-limit              0
anonymous-sdp                disabled
arp-msg-bandwidth            32000
rfc2833-timestamp            disabled
default-2833-duration        100
rfc2833-end-pkts-only-for-non-sig enabled
translate-non-rfc2833-event  disabled
media-supervision-traps      disabled
dnalg-server-failover        disabled
syslog-on-call-reject        disabled
last-modified-by             admin@IP
last-modified-date           2014-11-17 11:33:36
media-sec-policy
  name                        rtp
  pass-through                disabled
  options
  inbound
    profile
    mode                      rtp
    protocol                   none
  outbound
    profile
    mode                      rtp
    protocol                   none
  last-modified-by            admin@IP
  last-modified-date          2015-03-02 09:17:50
media-sec-policy
  name                        sfb-srtp
  pass-through                disabled
  options
  inbound
    profile                    sdes-profile
    mode                      srtp
    protocol                   sdes

```



```

outbound
  profile          sdes-profile
  mode            srtp
  protocol        sdes
  last-modified-by admin@IP
  last-modified-date 2015-03-02 09:17:30
network-interface
  name            s0p0
  sub-port-id    0
  description     For SIP-Trunk
  hostname
  ip-address      192.65.79.126
  pri-utility-addr
  sec-utility-addr
  netmask        255.255.255.224
  gateway        192.65.79.97
  sec-gateway
  gw-heartbeat
    state         disabled
    heartbeat     0
    retry-count   0
    retry-timeout 1
    health-score  0
  dns-ip-primary
  dns-ip-backup1
  dns-ip-backup2
  dns-domain
  dns-timeout    11
  signaling-mtu  0
  hip-ip-list    192.65.79.126
  ftp-address
  icmp-address   192.65.79.126
  snmp-address
  telnet-address
  ssh-address
  last-modified-by admin@IP
  last-modified-date 2015-09-02 10:06:42
network-interface
  name            s1p0
  sub-port-id    0
  description     Facing Skype for Business
  hostname        oracle.sfbldm.local
  ip-address      10.64.3.163
  pri-utility-addr
  sec-utility-addr
  netmask        255.255.0.0
  gateway        10.64.1.1
  sec-gateway
  gw-heartbeat
    state         disabled
    heartbeat     0
    retry-count   0
    retry-timeout 1
    health-score  0
  dns-ip-primary 172.16.29.42
  dns-ip-backup1

```

```

dns-ip-backup2
dns-domain                sfblabdm.local
dns-timeout                11
signaling-mtu              0
hip-ip-list                10.64.3.163
ftp-address
icmp-address               10.64.3.163
snmp-address
telnet-address
ssh-address                10.64.3.163
last-modified-by          admin@IP
last-modified-date        2015-09-11 14:51:42
network-interface
name                       wancom1
sub-port-id                0
description
hostname
ip-address
pri-utility-addr           169.254.1.1
sec-utility-addr           169.254.1.2
netmask                    255.255.255.252
gateway
sec-gateway
gw-heartbeat
    state                   disabled
    heartbeat                0
    retry-count              0
    retry-timeout            1
    health-score             0
dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout                11
signaling-mtu              0
hip-ip-list
ftp-address
icmp-address
snmp-address
telnet-address
ssh-address
last-modified-by          admin@IP
last-modified-date        2014-11-17 11:49:35
network-interface
name                       wancom2
sub-port-id                0
description
hostname
ip-address
pri-utility-addr           169.254.2.1
sec-utility-addr           169.254.2.2
netmask                    255.255.255.252
gateway
sec-gateway
gw-heartbeat
    state                   disabled

```

```

        heartbeat                0
        retry-count               0
        retry-timeout             1
        health-score              0

dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout                      11
signaling-mtu                    0
hip-ip-list
ftp-address
icmp-address
snmp-address
telnet-address
ssh-address
last-modified-by                 admin@IP
last-modified-date               2014-11-17 11:50:25
phy-interface
name                             s0p0
operation-type                   Media
port                             0
slot                             0
virtual-mac                      00:08:25:04:0d:1e
admin-state                      enabled
auto-negotiation                 enabled
duplex-mode                      FULL
speed                            100
wancom-health-score              50
overload-protection              disabled
last-modified-by                 admin@IP
last-modified-date               2014-11-17 11:53:04
phy-interface
name                             s1p0
operation-type                   Media
port                             0
slot                             1
virtual-mac                      00:08:25:04:0d:1f
admin-state                      enabled
auto-negotiation                 enabled
duplex-mode                      FULL
speed                            100
wancom-health-score              50
overload-protection              disabled
last-modified-by                 admin@IP
last-modified-date               2014-11-17 11:53:40
phy-interface
name                             wancom1
operation-type                   Control
port                             1
slot                             0
virtual-mac
admin-state                      enabled
auto-negotiation                 enabled
duplex-mode
speed

```

```

wancom-health-score      8
overload-protection      disabled
last-modified-by        admin@IP
last-modified-date      2014-11-17 11:54:41
phy-interface
  name                    wancom2
  operation-type          Control
  port                    2
  slot                    0
  virtual-mac
  admin-state             enabled
  auto-negotiation        enabled
  duplex-mode
  speed
  wancom-health-score    9
  overload-protection      disabled
  last-modified-by        admin@IP
  last-modified-date      2014-11-17 11:55:46
playback-config
  name                    transferrbt
  entry
    encoding               PCMU
    filename                US_ringbackPCMU.raw
    bytes-per-sec          8000
  last-modified-by        admin@IP
  last-modified-date      2014-11-17 12:04:41
realm-config
  identifier              SIP-Trunk
  description
  addr-prefix             0.0.0.0
  network-interfaces      s0p0:0
  mm-in-realm             enabled
  mm-in-network           enabled
  mm-same-ip              enabled
  mm-in-system            enabled
  bw-cac-non-mm          disabled
  msm-release             disabled
  qos-enable              disabled
  max-bandwidth           0
  fallback-bandwidth      0
  max-priority-bandwidth  0
  max-latency             0
  max-jitter              0
  max-packet-loss         0
  observ-window-size      0
  parent-realm
  dns-realm
  media-policy
  media-sec-policy        rtp
  srtp-msm-passthrough    disabled
  class-profile
  in-translationid
  out-translationid
  in-manipulationid
  out-manipulationid
  average-rate-limit      0

```

access-control-trust-level	none
invalid-signal-threshold	0
maximum-signal-threshold	0
untrusted-signal-threshold	0
nat-trust-threshold	0
max-endpoints-per-nat	0
nat-invalid-message-threshold	0
wait-time-for-invalid-register	0
deny-period	30
cac-failure-threshold	0
untrust-cac-failure-threshold	0
ext-policy-svr	
diam-e2-address-realm	
subscription-id-type	END_USER_NONE
symmetric-latching	disabled
pai-strip	disabled
trunk-context	
device-id	
early-media-allow	
enforcement-profile	
additional-prefixes	
restricted-latching	none
restriction-mask	32
user-cac-mode	none
user-cac-bandwidth	0
user-cac-sessions	0
icmp-detect-multiplier	0
icmp-advertisement-interval	0
icmp-target-ip	
monthly-minutes	0
options	
spl-options	
accounting-enable	enabled
net-management-control	disabled
delay-media-update	disabled
refer-call-transfer	disabled
hold-refer-reinvite	disabled
refer-notify-provisional	none
dyn-refer-term	disabled
codec-policy	
codec-manip-in-realm	disabled
codec-manip-in-network	enabled
rtcp-policy	
constraint-name	
session-recording-server	
session-recording-required	disabled
manipulation-string	
manipulation-pattern	
stun-enable	disabled
stun-server-ip	0.0.0.0
stun-server-port	3478
stun-changed-ip	0.0.0.0
stun-changed-port	3479
sip-profile	
sip-isup-profile	
match-media-profiles	

```

qos-constraint
block-rtcp                disabled
hide-egress-media-update  disabled
tcp-media-profile
monitoring-filters
node-functionality
default-location-string
alt-family-realm
pref-addr-type            none
last-modified-by         admin@IP
last-modified-date       2015-09-18 15:37:41
realm-config
  identifier               towards-sfb
  description
  addr-prefix              0.0.0.0
  network-interfaces       slp0:0
  mm-in-realm              enabled
  mm-in-network            enabled
  mm-same-ip               enabled
  mm-in-system             enabled
  bw-cac-non-mm           disabled
  msm-release              disabled
  qos-enable               disabled
  max-bandwidth            0
  fallback-bandwidth       0
  max-priority-bandwidth  0
  max-latency              0
  max-jitter               0
  max-packet-loss          0
  observ-window-size       0
  parent-realm
  dns-realm
  media-policy
  media-sec-policy         sfb-srtp
  srtp-msm-passthrough     disabled
  class-profile
  in-translationid
  out-translationid
  in-manipulationid
  out-manipulationid
  average-rate-limit       0
  access-control-trust-level none
  invalid-signal-threshold 0
  maximum-signal-threshold 0
  untrusted-signal-threshold 0
  nat-trust-threshold      0
  max-endpoints-per-nat    0
  nat-invalid-message-threshold 0
  wait-time-for-invalid-register 0
  deny-period              30
  cac-failure-threshold    0
  untrust-cac-failure-threshold 0
  ext-policy-svr
  diam-e2-address-realm
  subscription-id-type     END_USER_NONE
  symmetric-latching       disabled

```

pai-strip	disabled
trunk-context	
device-id	
early-media-allow	
enforcement-profile	
additional-prefixes	
restricted-latching	none
restriction-mask	32
user-cac-mode	none
user-cac-bandwidth	0
user-cac-sessions	0
icmp-detect-multiplier	0
icmp-advertisement-interval	0
icmp-target-ip	
monthly-minutes	0
options	
spl-options	
accounting-enable	enabled
net-management-control	disabled
delay-media-update	disabled
refer-call-transfer	enabled
hold-refer-reinvite	disabled
refer-notify-provisional	none
dyn-refer-term	disabled
codec-policy	
codec-manip-in-realm	disabled
codec-manip-in-network	enabled
rtcp-policy	
constraint-name	
session-recording-server	
session-recording-required	disabled
manipulation-string	
manipulation-pattern	
stun-enable	disabled
stun-server-ip	0.0.0.0
stun-server-port	3478
stun-changed-ip	0.0.0.0
stun-changed-port	3479
sip-profile	
sip-isup-profile	
match-media-profiles	
qos-constraint	
block-rtcp	disabled
hide-egress-media-update	disabled
tcp-media-profile	
monitoring-filters	
node-functionality	
default-location-string	
alt-family-realm	
pref-addr-type	none
last-modified-by	admin@IP
last-modified-date	2015-09-18 15:38:06
redundancy-config	
state	enabled
log-level	INFO
health-threshold	75

```

emergency-threshold      50
port                    9090
advertisement-time      500
percent-drift            210
initial-time             1250
becoming-standby-time   180000
becoming-active-time    100
cfg-port                 1987
cfg-max-trans            10000
cfg-sync-start-time     5000
cfg-sync-comp-time      1000
gateway-heartbeat-interval 0
gateway-heartbeat-retry 0
gateway-heartbeat-timeout 1
gateway-heartbeat-health 0
media-if-peercheck-time 0
peer
  name                   Acmesystem1
  state                   enabled
  type                    Primary
  destination
    address                169.254.1.1:9090
    network-interface      wancom1:0
  destination
    address                169.254.2.1:9090
    network-interface      wancom2:0
peer
  name                   acmesystem3
  state                   enabled
  type                    Secondary
  destination
    address                169.254.1.2:9090
    network-interface      wancom1:0
  destination
    address                169.254.2.2:9090
    network-interface      wancom2:0
last-modified-by        admin@IP
last-modified-date      2014-11-17 12:27:59
response-map
  name                   change183to180
  entries
    recv-code             183
    xmit-code             180
    reason                 Ringing
    method
    register-response-expires
  last-modified-by        admin@IP
  last-modified-date      2015-02-20 10:37:37
sdes-profile
  name                   sdes-profile
  crypto-list             AES_CM_128_HMAC_SHA1_80
  srtp-auth               enabled
  srtp-encrypt            enabled
  srtcp-encrypt           enabled
  mki                     disabled
  egress-offer-format     same-as-ingress

```



```

use-ingress-session-params
options
key
salt
srtp-rekey-on-re-invite          disabled
last-modified-by                admin@IP
last-modified-date              2015-03-04 11:18:36
session-agent
hostname                        63.87.147.48
ip-address                      63.87.147.48
port                            5071
state                           enabled
app-protocol                    SIP
app-type
transport-method                UDP
realm-id                        SIP-Trunk
egress-realm-id
description
carriers
allow-next-hop-lp               enabled
constraints                     disabled
max-sessions                    0
max-inbound-sessions            0
max-outbound-sessions          0
max-burst-rate                  0
max-inbound-burst-rate          0
max-outbound-burst-rate        0
max-sustain-rate                0
max-inbound-sustain-rate        0
max-outbound-sustain-rate       0
min-seizures                    5
min-asr                         0
time-to-resume                  0
ttr-no-response                 0
in-service-period               0
burst-rate-window               0
sustain-rate-window             0
req-uri-carrier-mode            None
proxy-mode
redirect-action
loose-routing                   enabled
send-media-session              enabled
response-map
ping-method                     OPTIONS
ping-interval                   30
ping-send-mode                  keep-alive
ping-all-addresses             disabled
ping-in-service-response-codes
out-service-response-codes
load-balance-dns-query          hunt
options
spl-options
media-profiles
in-translationid
out-translationid               stripplus1
trust-me                        disabled

```

```

request-uri-headers
stop-recurse
local-response-map
ping-to-user-part
ping-from-user-part
in-manipulationid
out-manipulationid
manipulation-string
manipulation-pattern
p-asserted-id
trunk-group
max-register-sustain-rate          0
early-media-allow
invalidate-registrations            disabled
rfc2833-mode                        none
rfc2833-payload                     0
codec-policy
enforcement-profile
refer-call-transfer                 disabled
refer-notify-provisional            none
reuse-connections                   NONE
tcp-keepalive                       none
tcp-reconn-interval                0
max-register-burst-rate             0
register-burst-window               0
sip-profile
sip-isup-profile
kpml-interworking                   inherit
monitoring-filters
session-recording-server
session-recording-required          disabled
hold-refer-reinvite                 disabled
send-tcp-fin                        disabled
last-modified-by                    web_admin@IP
last-modified-date                  2015-09-17 12:54:39
session-agent
hostname                             Medpool.sfbldm.local
ip-address
port                                 5067
state                                 enabled
app-protocol                         SIP
app-type
transport-method                    StaticTLS
realm-id                             towards-sfb
egress-realm-id
description
carriers
allow-next-hop-lp                   enabled
constraints                          disabled
max-sessions                         0
max-inbound-sessions                 0
max-outbound-sessions               0
max-burst-rate                      0
max-inbound-burst-rate              0
max-outbound-burst-rate             0
max-sustain-rate                    0

```

max-inbound-sustain-rate	0
max-outbound-sustain-rate	0
min-seizures	5
min-asr	0
time-to-resume	0
ttr-no-response	0
in-service-period	0
burst-rate-window	0
sustain-rate-window	0
req-uri-carrier-mode	None
proxy-mode	
redirect-action	
loose-routing	enabled
send-media-session	enabled
response-map	
ping-method	OPTIONS
ping-interval	60
ping-send-mode	keep-alive
ping-all-addresses	disabled
ping-in-service-response-codes	
out-service-response-codes	
load-balance-dns-query	hunt
options	
spl-options	
media-profiles	
in-translationid	
out-translationid	addplus1
trust-me	disabled
request-uri-headers	
stop-recurse	480
local-response-map	
ping-to-user-part	
ping-from-user-part	
in-manipulationid	
out-manipulationid	
manipulation-string	
manipulation-pattern	
p-asserted-id	
trunk-group	
max-register-sustain-rate	0
early-media-allow	
invalidate-registrations	disabled
rfc2833-mode	none
rfc2833-payload	0
codec-policy	
enforcement-profile	
refer-call-transfer	enabled
refer-notify-provisional	none
reuse-connections	NONE
tcp-keepalive	disabled
tcp-reconn-interval	0
max-register-burst-rate	0
register-burst-window	0
sip-profile	
sip-isup-profile	
kpml-interworking	inherit

```

monitoring-filters
session-recording-server
session-recording-required          disabled
hold-refer-reinvite                 disabled
send-tcp-fin                         disabled
last-modified-by                    web_admin@IP
last-modified-date                   2015-09-21 11:03:47
session-agent
hostname                             med1.sfbblabdm.local
ip-address                           172.16.29.44
port                                 5067
state                                 enabled
app-protocol                         SIP
app-type
transport-method                     StaticTLS
realm-id                             towards-sfb
egress-realm-id
description
carriers
allow-next-hop-lp                    enabled
constraints                           disabled
max-sessions                          0
max-inbound-sessions                 0
max-outbound-sessions                0
max-burst-rate                       0
max-inbound-burst-rate               0
max-outbound-burst-rate              0
max-sustain-rate                     0
max-inbound-sustain-rate             0
max-outbound-sustain-rate            0
min-seizures                          5
min-asr                              0
time-to-resume                       0
ttr-no-response                      0
in-service-period                    0
burst-rate-window                    0
sustain-rate-window                  0
req-uri-carrier-mode                 None
proxy-mode
redirect-action
loose-routing                         enabled
send-media-session                   enabled
response-map
ping-method                           OPTIONS
ping-interval                         60
ping-send-mode                        keep-alive
ping-all-addresses                   disabled
ping-in-service-response-codes
out-service-response-codes
load-balance-dns-query                hunt
options
spl-options
media-profiles
in-translationid
out-translationid                     addplus1
trust-me                              disabled

```

```

request-uri-headers
stop-recurse
local-response-map
ping-to-user-part
ping-from-user-part
in-manipulationid
out-manipulationid
manipulation-string
manipulation-pattern
p-asserted-id
trunk-group
max-register-sustain-rate          0
early-media-allow
invalidate-registrations            disabled
rfc2833-mode                        none
rfc2833-payload                     0
codec-policy
enforcement-profile
refer-call-transfer                 enabled
refer-notify-provisional            none
reuse-connections                   NONE
tcp-keepalive                       none
tcp-reconn-interval                0
max-register-burst-rate             0
register-burst-window               0
sip-profile
sip-isup-profile
kpml-interworking                   inherit
monitoring-filters
session-recording-server
session-recording-required          disabled
hold-refer-reinvite                 disabled
send-tcp-fin                        disabled
last-modified-by                    admin@IP
last-modified-date                  2015-09-21 12:56:40
session-agent
hostname                             med2.sfblabdm.local
ip-address                          172.16.29.45
port                                 5067
state                               enabled
app-protocol                        SIP
app-type
transport-method                    StaticTLS
realm-id                             towards-sfb
egress-realm-id
description
carriers
allow-next-hop-lp                   enabled
constraints                          disabled
max-sessions                         0
max-inbound-sessions                 0
max-outbound-sessions               0
max-burst-rate                      0
max-inbound-burst-rate              0
max-outbound-burst-rate             0
max-sustain-rate                    0

```

max-inbound-sustain-rate	0
max-outbound-sustain-rate	0
min-seizures	5
min-asr	0
time-to-resume	0
ttr-no-response	0
in-service-period	0
burst-rate-window	0
sustain-rate-window	0
req-uri-carrier-mode	None
proxy-mode	
redirect-action	
loose-routing	enabled
send-media-session	enabled
response-map	
ping-method	OPTIONS
ping-interval	60
ping-send-mode	keep-alive
ping-all-addresses	disabled
ping-in-service-response-codes	
out-service-response-codes	
load-balance-dns-query	hunt
options	
spl-options	
media-profiles	
in-translationid	
out-translationid	addplus1
trust-me	disabled
request-uri-headers	
stop-recurse	
local-response-map	
ping-to-user-part	
ping-from-user-part	
in-manipulationid	
out-manipulationid	
manipulation-string	
manipulation-pattern	
p-asserted-id	
trunk-group	
max-register-sustain-rate	0
early-media-allow	
invalidate-registrations	disabled
rfc2833-mode	none
rfc2833-payload	0
codec-policy	
enforcement-profile	
refer-call-transfer	enabled
refer-notify-provisional	none
reuse-connections	NONE
tcp-keepalive	none
tcp-reconn-interval	0
max-register-burst-rate	0
register-burst-window	0
sip-profile	
sip-isup-profile	
kpml-interworking	inherit

```

monitoring-filters
session-recording-server
session-recording-required          disabled
hold-refer-reinvite                 disabled
send-tcp-fin                         disabled
last-modified-by                    admin@IP
last-modified-date                   2015-09-21 12:57:06
session-translation
  id                                  addplus1
  rules-calling                       addplus1
  rules-called                        addplus1
  last-modified-by                    admin@IP
  last-modified-date                  2015-03-05 12:14:27
session-translation
  id                                  stripplus1
  rules-calling                       stripplus1
  rules-called                        stripplus1
  last-modified-by                    admin@10.64.204.12
  last-modified-date                  2015-08-28 12:00:43
sip-config
  state                               enabled
  operation-mode                      dialog
  dialog-transparency                 enabled
  home-realm-id                       towards-sfb
  egress-realm-id
  auto-realm-id
  nat-mode                            None
  registrar-domain                    *
  registrar-host                      *
  registrar-port                      0
  register-service-route              always
  init-timer                          500
  max-timer                           4000
  trans-expire                        32
  initial-inv-trans-expire            0
  invite-expire                       180
  inactive-dynamic-conn               32
  enforcement-profile
  pac-method
  pac-interval                        10
  pac-strategy                        PropDist
  pac-load-weight                     1
  pac-session-weight                  1
  pac-route-weight                    1
  pac-callid-lifetime                 600
  pac-user-lifetime                   3600
  red-sip-port                        1988
  red-max-trans                       10000
  red-sync-start-time                 5000
  red-sync-comp-time                  1000
  options                             max-udp-length=0
  add-reason-header                   disabled
  sip-message-len                     4096
  enum-sag-match                      disabled
  extra-method-stats                  disabled
  extra-enum-stats                    disabled

```

```

rph-feature disabled
nsep-user-sessions-rate 0
nsep-sa-sessions-rate 0
registration-cache-limit 0
register-use-to-for-lp disabled
refer-src-routing disabled
add-ucid-header disabled
proxy-sub-events
allow-pani-for-trusted-only disabled
atcf-stn-sr
atcf-psi-dn
atcf-route-to-sccas disabled
eatf-stn-sr
pass-gruu-contact disabled
sag-lookup-on-redirect disabled
set-disconnect-time-on-bye disabled
msrp-delayed-bye-timer 15
transcoding-realm
transcoding-agents
create-dynamic-sa disabled
node-functionality P-CSCF
match-sip-instance disabled
sa-routes-stats disabled
sa-routes-traps disabled
rx-sip-reason-mapping disabled
add-ue-location-in-pani disabled
hold-emergency-calls-for-loc-info 0
last-modified-by web_admin@IP
last-modified-date 2015-09-17 12:42:41
sip-feature
name 100rel
realm SIP-Trunk
support-mode-inbound Pass
require-mode-inbound Pass
proxy-require-mode-inbound Pass
support-mode-outbound Pass
require-mode-outbound Pass
proxy-require-mode-outbound Pass
last-modified-by admin@10.64.204.12
last-modified-date 2015-08-28 14:56:44
sip-interface
state enabled
realm-id SIP-Trunk
description
sip-port
address 192.65.79.126
port 5060
transport-protocol UDP
tls-profile
allow-anonymous all
multi-home-addr
ims-aka-profile
carriers
trans-expire 0
initial-inv-trans-expire 0
invite-expire 0

```


max-redirect-contacts	0
proxy-mode	
redirect-action	
contact-mode	none
nat-traversal	none
nat-interval	30
tcp-nat-interval	90
registration-caching	disabled
min-reg-expire	300
registration-interval	3600
route-to-registrar	disabled
secured-network	disabled
teluri-scheme	disabled
uri-fqdn-domain	
options	
spl-options	
trust-mode	all
max-nat-interval	3600
nat-int-increment	10
nat-test-increment	30
sip-dynamic-hnt	disabled
stop-recurse	401,407
port-map-start	0
port-map-end	0
in-manipulationid	
out-manipulationid	NATandChangeContact
sip-ims-feature	disabled
sip-atcf-feature	disabled
subscribe-reg-event	disabled
operator-identifier	
anonymous-priority	none
max-incoming-conns	0
per-src-ip-max-incoming-conns	0
inactive-conn-timeout	0
untrusted-conn-timeout	0
network-id	
ext-policy-server	
ldap-policy-server	
default-location-string	
term-tgrp-mode	none
charging-vector-mode	pass
charging-function-address-mode	pass
ccf-address	
ecf-address	
implicit-service-route	disabled
rfc2833-payload	101
rfc2833-mode	transparent
constraint-name	
response-map	
local-response-map	
sec-agree-feature	disabled
sec-agree-pref	ipsec3gpp
enforcement-profile	
route-unauthorized-calls	
tcp-keepalive	none
add-sdp-invite	disabled

```

p-early-media-header          disabled
p-early-media-direction
add-sdp-profiles
manipulation-string
manipulation-pattern
sip-profile
sip-isup-profile
tcp-conn-dereg                0
tunnel-name
register-keep-alive           none
kpml-interworking             disabled
msrp-delay-egress-bye        disabled
send-380-response
pcscf-restoration
session-timer-profile
session-recording-server
session-recording-required    disabled
service-tag
reg-cache-route               disabled
last-modified-by              admin@IP
last-modified-date            2015-09-08 17:49:21
sip-interface
state                          enabled
realm-id                       towards-sfb
description
sip-port
    address                     10.64.3.163
    port                         5067
    transport-protocol           TLS
    tls-profile                   sfb-tls-profile
    allow-anonymous               all
    multi-home-addr
    ims-aka-profile
carriers
trans-expire                   0
initial-inv-trans-expire       0
invite-expire                  0
max-redirect-contacts          0
proxy-mode
redirect-action
contact-mode                   none
nat-traversal                  none
nat-interval                   30
tcp-nat-interval               90
registration-caching           disabled
min-reg-expire                 300
registration-interval          3600
route-to-registrar             disabled
secured-network                disabled
teluri-scheme                  disabled
uri-fqdn-domain
options                        100rel-interworking
spl-options
trust-mode                     all
max-nat-interval               3600
nat-int-increment              10

```

nat-test-increment	30
sip-dynamic-hnt	disabled
stop-recurse	401,407
port-map-start	0
port-map-end	0
in-manipulationid	ForEarlyMedia
out-manipulationid	outManipToSFB
sip-ims-feature	disabled
sip-atcf-feature	disabled
subscribe-reg-event	disabled
operator-identifier	
anonymous-priority	none
max-incoming-conns	0
per-src-ip-max-incoming-conns	0
inactive-conn-timeout	0
untrusted-conn-timeout	0
network-id	
ext-policy-server	
ldap-policy-server	
default-location-string	
term-tgrp-mode	none
charging-vector-mode	pass
charging-function-address-mode	pass
ccf-address	
ecf-address	
implicit-service-route	disabled
rfc2833-payload	101
rfc2833-mode	transparent
constraint-name	
response-map	
local-response-map	
sec-agree-feature	disabled
sec-agree-pref	ipsec3gpp
enforcement-profile	
route-unauthorized-calls	
tcp-keepalive	none
add-sdp-invite	disabled
p-early-media-header	disabled
p-early-media-direction	
add-sdp-profiles	
manipulation-string	
manipulation-pattern	
sip-profile	
sip-isup-profile	
tcp-conn-dereg	0
tunnel-name	
register-keep-alive	none
kpml-interworking	disabled
msrp-delay-egress-bye	disabled
send-380-response	
pcscf-restoration	
session-timer-profile	
session-recording-server	
session-recording-required	disabled
service-tag	
reg-cache-route	disabled

```

last-modified-by      admin@IP
last-modified-date    2015-09-18 15:21:36
sip-manipulation
  name                 Addpcmato183
  description
  split-headers
  join-headers
  header-rule
    name               check1for83
    header-name        @status-line
    action              manipulate
    comparison-type    pattern-rule
    msg-type           any
    methods
    match-value
    new-value
    element-rule
      name              isit183
      parameter-name
      type              status-code
      action            sip-manip
      match-val-type   any
      comparison-type  pattern-rule
      match-value      183
      new-value        fix183sdp
  last-modified-by    admin@IP
  last-modified-date  2015-09-10 12:26:47
sip-manipulation
  name                 ChangeContact
  description
  split-headers
  join-headers
  header-rule
    name               StoreFromnumber
    header-name        From
    action              manipulate
    comparison-type    case-sensitive
    msg-type           any
    methods
    match-value
    new-value
    element-rule
      name              StoreFromnumber_er
      parameter-name
      type              uri-user-only
      action            store
      match-val-type   any
      comparison-type  case-sensitive
      match-value
      new-value
  header-rule
    name               ChangeContact
    header-name        Contact
    action              manipulate
    comparison-type    case-sensitive
    msg-type           any

```

```

        methods
        match-value
        new-value
        element-rule
            name                ChangeContact_er
            parameter-name
            type                  uri-user
            action                 add
            match-val-type        any
            comparison-type       case-sensitive
            match-value
            new-value
$StoreFromnumber.$StoreFromnumber_er.$0
    last-modified-by            admin@IP
    last-modified-date          2014-11-17 12:50:13
sip-manipulation
    name                         ChangeRecvOnlyToInactive
    description
    split-headers
    join-headers
    header-rule
        name                     changeSDP
        header-name               Content-Type
        action                    manipulate
        comparison-type           case-sensitive
        msg-type                  reply
        methods                   INVITE
        match-value
        new-value
        element-rule
            name                   RecvOnlyToInactive
            parameter-name         application/sdp
            type                   mime
            action                 find-replace-all
            match-val-type        any
            comparison-type       pattern-rule
            match-value           a=recvonly
            new-value             a=inactive
    last-modified-by            admin@IP
    last-modified-date          2015-02-26 12:59:33
sip-manipulation
    name                         Changeinactosendonly
    description                   Change inactive to sendonly for pstn tran
    split-headers
    join-headers
    header-rule
        name                     changeSDP
        header-name               Content-Type
        action                    manipulate
        comparison-type           case-sensitive
        msg-type                  request
        methods                   INVITE
        match-value
        new-value
        element-rule
            name                   inacttosendonly

```

```

parameter-name      application/sdp
type                mime
action              find-replace-all
match-val-type      any
comparison-type     pattern-rule
match-value         a=inactive
new-value           a=sendonly

last-modified-by    admin@IP
last-modified-date  2014-11-17 12:53:24
sip-manipulation
name                Check183
description
split-headers
join-headers
header-rule
  name              check183
  header-name       @status-line
  action            manipulate
  comparison-type   pattern-rule
  msg-type          any
  methods
  match-value
  new-value
  element-rule
    name            is183
    parameter-name
    type            status-code
    action          sip-manip
    match-val-type  any
    comparison-type pattern-rule
    match-value     183
    new-value       Modcline

last-modified-by    admin@IP
last-modified-date  2015-09-09 15:12:56
sip-manipulation
name                ForEarlyMedia
description
split-headers
join-headers
header-rule
  name              delsupported
  header-name       Supported
  action            delete
  comparison-type   case-sensitive
  msg-type          request
  methods           INVITE
  match-value
  new-value

header-rule
  name              addrequireinINVITE
  header-name       Require
  action            add
  comparison-type   case-sensitive
  msg-type          request
  methods           INVITE
  match-value

```

```

        new-value                100rel
    header-rule
        name                      Options
        header-name                From
        action                     sip-manip
        comparison-type            case-sensitive
        msg-type                   any
        methods
        match-value
        new-value                  OptionsResponseLocally
    header-rule
        name                      Change183
        header-name                From
        action                     sip-manip
        comparison-type            case-sensitive
        msg-type                   any
        methods
        match-value
        new-value                  Check183
    last-modified-by              admin@IP
    last-modified-date            2015-09-09 12:13:23
sip-manipulation
    name                          Modcline
    description
    split-headers
    join-headers
    mime-sdp-rule
        name                      test
        msg-type                   any
        methods
        action                     manipulate
        comparison-type            pattern-rule
        match-value
        new-value
        sdp-media-rule
            name                   checkcline
            media-type              audio
            action                  manipulate
            comparison-type          pattern-rule
            match-value
            new-value
            sdp-line-rule
                name                detectc
                type                 c
                action              sip-manip
                comparison-type      pattern-rule
                match-value
                new-value
                ^(.?!(172.16.29.44|172.16.29.45|10.64.3.163))*$
                new-value            convert183to180
    header-rule
        name                      checkcl
        header-name                Content-Length
        action                     sip-manip
        comparison-type            case-sensitive
        msg-type                   any
        methods

```

```

        match-value          0
        new-value            convert183to180
last-modified-by          web_admin@IP
last-modified-date       2015-09-09 18:26:21
sip-manipulation
  name                     NATandChangeContact
  description              NAT plus Change Contact
  split-headers
  join-headers
  header-rule
    name                   doNATfortrunk
    header-name            From
    action                 sip-manip
    comparison-type        case-sensitive
    msg-type               any
    methods
    match-value
    new-value              NATting
  header-rule
    name                   delsupported
    header-name            Supported
    action                 delete
    comparison-type        case-sensitive
    msg-type               request
    methods                INVITE
    match-value
    new-value
  header-rule
    name                   addPAIonCallForwards
    header-name            From
    action                 sip-manip
    comparison-type        case-sensitive
    msg-type               request
    methods                INVITE
    match-value
    new-value              addPAIonCallForwards
  header-rule
    name                   addPAIRefers
    header-name            From
    action                 sip-manip
    comparison-type        case-sensitive
    msg-type               request
    methods                INVITE
    match-value
    new-value              addPAIRefers
  header-rule
    name                   ChangeContactHeader
    header-name            From
    action                 sip-manip
    comparison-type        case-sensitive
    msg-type               any
    methods
    match-value
    new-value              ChangeContact
  header-rule
    name                   modify183sdp

```



```

        header-name          From
        action                sip-manip
        comparison-type       case-sensitive
        msg-type              any
        methods
        match-value
        new-value              Addpcmat0183
    last-modified-by         web@
    last-modified-date       2015-09-10 14:48:23
sip-manipulation
    name                      NATting
    description
    split-headers
    join-headers
    header-rule
        name                  From
        header-name           From
        action                 manipulate
        comparison-type        case-sensitive
        msg-type               request
        methods
        match-value
        new-value
        element-rule
            name                From_header
            parameter-name
            type                 uri-host
            action               replace
            match-val-type       any
            comparison-type      case-sensitive
            match-value
            new-value            $LOCAL_IP
    header-rule
        name                  To
        header-name           To
        action                 manipulate
        comparison-type        case-sensitive
        msg-type               request
        methods
        match-value
        new-value
        element-rule
            name                To
            parameter-name
            type                 uri-host
            action               replace
            match-val-type       any
            comparison-type      case-sensitive
            match-value
            new-value            $REMOTE_IP
    last-modified-by         web_admin@IP
    last-modified-date       2015-09-10 16:38:21
sip-manipulation
    name                      OptionsResponseLocally
    description
    split-headers

```

```

join-headers
header-rule
    name rejectOptions
    header-name request-uri
    action reject
    comparison-type case-sensitive
    msg-type request
    methods OPTIONS
    match-value
    new-value 200: OK
last-modified-by admin@IP
last-modified-date 2015-09-02 11:29:29
sip-manipulation
    name addPAIRefers
    description
    split-headers
    join-headers
    header-rule
        name checkReferredBy
        header-name REFERRED-BY
        action store
        comparison-type case-sensitive
        msg-type request
        methods INVITE
        match-value
        new-value
    header-rule
        name addPAIifReferredBy
        header-name P-Asserted-Identity
        action add
        comparison-type pattern-rule
        msg-type request
        methods INVITE
        match-value $checkReferredBy
        new-value <sip:5712935325+@+$LOCAL_IP+>
last-modified-by admin@IP
last-modified-date 2015-02-26 10:10:30
sip-manipulation
    name addPAIonCallForwards
    description
    split-headers
    join-headers
    header-rule
        name checkForHistoryInfo
        header-name History-Info
        action store
        comparison-type case-sensitive
        msg-type request
        methods INVITE
        match-value
        new-value
    header-rule
        name addPAIifHistoryInfo
        header-name P-Asserted-Identity
        action add
        comparison-type pattern-rule

```

```

msg-type request
methods INVITE
match-value $checkForHistoryInfo
new-value <sip:5712935325+@+$LOCAL_IP+>
last-modified-by admin@IP
last-modified-date 2015-02-20 12:06:39
sip-manipulation
name convert183to180
description
split-headers
join-headers
header-rule
name delSDP
header-name Content-Type
action manipulate
comparison-type case-insensitive
msg-type any
methods
match-value
new-value
element-rule
name del183SDP
parameter-name application/sdp
type mime
action delete-element
match-val-type any
comparison-type case-sensitive
match-value
new-value
header-rule
name delContentType
header-name Content-Type
action manipulate
comparison-type case-sensitive
msg-type any
methods
match-value
new-value
element-rule
name delCT
parameter-name *
type header-param
action delete-header
match-val-type any
comparison-type case-sensitive
match-value
new-value
header-rule
name change487to486
header-name @status-line
action manipulate
comparison-type case-sensitive
msg-type any
methods
match-value
new-value

```

```

        element-rule
            name                               modStatusCode
            parameter-name
            type                               status-code
            action                             replace
            match-val-type                    any
            comparison-type                   case-sensitive
            match-value                       183
            new-value                         180
        element-rule
            name                               modReasonPhrase
            parameter-name
            type                               reason-phrase
            action                             replace
            match-val-type                    any
            comparison-type                   case-insensitive
            match-value                       Session Progress
            new-value                         Ringing
    last-modified-by                       admin@IP
    last-modified-date                     2015-09-09 11:59:41
sip-manipulation
    name                                   fix183sdp
    description                           To bypass the reinvoke
    split-headers
    join-headers
    header-rule
        name                               Checkforpcma
        header-name                       Content-type
        action                             store
        comparison-type                   case-sensitive
        msg-type                           any
        methods
        match-value
        new-value
    element-rule
        name                               Checkpcmaexists
        parameter-name                   application/sdp
        type                               mime
        action                             store
        match-val-type                    any
        comparison-type                   case-sensitive
        match-value                       (a=rtpmap:8 PCMA/8000)
        new-value
    header-rule
        name                               Addpcma
        header-name                       Content-Type
        action                             manipulate
        comparison-type                   boolean
        msg-type                           any
        match-value                       !$Checkforpcma.$Checkpcmaexists
        new-value
    element-rule
        name                               Addpcma
        parameter-name                   application/sdp
        type                               mime

```

```

        action find-replace-all
        match-val-type any
        comparison-type pattern-rule
        match-value a=rtpmap:0 PCMU/8000
        new-value "a=rtpmap:0
PCMU/8000"+$CRLF+"a=rtpmap:8 PCMA/8000"
    mime-sdp-rule
        name modmline
        msg-type any
        methods
        action manipulate
        comparison-type case-sensitive
        match-value
        new-value
        sdp-media-rule
            name modmline_m
            media-type audio
            action manipulate
            comparison-type case-sensitive
            match-value
            new-value
            sdp-line-rule
                name change_payload
                type m
                action find-replace-all
                comparison-type pattern-rule
                match-value ^(audio) ( [0-
9]{4,5}) ( RTP/AVP 0 101)$
                new-value audio+$2+" RTP/AVP 0 8
101"
        last-modified-by web_admin@IP
        last-modified-date 2015-09-10 13:02:42
    sip-manipulation
        name modcontact
        description
        split-headers
        join-headers
        header-rule
            name modcontact_er
            header-name Contact
            action manipulate
            comparison-type pattern-rule
            msg-type any
            methods
            match-value
            new-value
            element-rule
                name moder
                parameter-name
                type uri-user
                action replace
                match-val-type any
                comparison-type pattern-rule
                match-value ((\+1) (\+1\d+))
                new-value $3
            element-rule

```

```

        name modanon
        parameter-name
        type uri-user
        action replace
        match-val-type any
        comparison-type pattern-rule
        match-value \+lanonymous
        new-value anonymous
    last-modified-by admin@IP
    last-modified-date 2015-09-15 12:01:44
sip-manipulation
    name outManipToSFB
    description
    split-headers
    join-headers
    header-rule
        name From
        header-name From
        action manipulate
        comparison-type case-sensitive
        msg-type request
        methods
        match-value
        new-value
        element-rule
            name From_header
            parameter-name
            type uri-host
            action replace
            match-val-type any
            comparison-type case-sensitive
            match-value
            new-value oracle.sfblabdm.local
        element-rule
            name modanonfrom
            parameter-name
            type uri-user
            action replace
            match-val-type any
            comparison-type pattern-rule
            match-value \+lanonymous
            new-value anonymous
    header-rule
        name To
        header-name To
        action manipulate
        comparison-type case-sensitive
        msg-type request
        methods
        match-value
        new-value
        element-rule
            name remplusone
            parameter-name
            type uri-user
            action replace

```

	match-val-type	any
	comparison-type	pattern-rule
	match-value	\+1\$
	new-value	
element-rule		
	name	To
	parameter-name	
	type	uri-host
	action	replace
	match-val-type	any
	comparison-type	case-sensitive
	match-value	
	new-value	medpool.sfblabdm.local
header-rule		
	name	fixcontact
	header-name	Contact
	action	manipulate
	comparison-type	case-sensitive
	msg-type	any
	methods	
	match-value	
	new-value	
element-rule		
	name	updatecontact
	parameter-name	
	type	uri-host
	action	replace
	match-val-type	any
	comparison-type	case-sensitive
	match-value	
	new-value	oracle.sfblabdm.local
header-rule		
	name	fixcontactuser
	header-name	Contact
	action	manipulate
	comparison-type	pattern-rule
	msg-type	request
	methods	INVITE
	match-value	
	new-value	
element-rule		
	name	user
	parameter-name	
	type	uri-user
	action	replace
	match-val-type	any
	comparison-type	pattern-rule
	match-value	(.*)
	new-value	"+1"+\$ORIGINAL
element-rule		
	name	moder
	parameter-name	
	type	uri-user
	action	replace
	match-val-type	any
	comparison-type	pattern-rule

match-value	((\+1) (\+1\d+))
new-value	\$3
element-rule	
name	modanoncon
parameter-name	
type	uri-user
action	replace
match-val-type	any
comparison-type	pattern-rule
match-value	\+1anonymous
new-value	anonymous
header-rule	
name	Ruri
header-name	Request-URI
action	manipulate
comparison-type	case-sensitive
msg-type	request
methods	
match-value	
new-value	
element-rule	
name	remplusone
parameter-name	
type	uri-user
action	replace
match-val-type	any
comparison-type	pattern-rule
match-value	\+1\$
new-value	
header-rule	
name	addReplaces
header-name	Supported
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
match-value	
new-value	\$ORIGINAL+, replaces
header-rule	
name	addUpdate
header-name	Allow
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
match-value	
new-value	\$ORIGINAL+, UPDATE
header-rule	
name	addSessionExpires
header-name	Session-Expires
action	add
comparison-type	case-sensitive
msg-type	request
methods	INVITE
match-value	
new-value	300


```

header-rule
  name addingptime
  header-name From
  action sip-manip
  comparison-type case-sensitive
  msg-type request
  methods
  match-value
  new-value replacemptimewithptime
last-modified-by admin@IP
last-modified-date 2015-09-15 15:12:01
sip-manipulation
  name replacemptimewithptime
  description
  split-headers
  join-headers
  mime-sdp-rule
    name addptime
    msg-type request
    methods
    action manipulate
    comparison-type case-sensitive
    match-value
    new-value
  sdp-media-rule
    name addSDPptime
    media-type audio
    action manipulate
    comparison-type case-sensitive
    match-value
    new-value
  sdp-line-rule
    name addptimelr
    type a
    action find-replace-all
    comparison-type case-sensitive
    match-value maxptime:20
    new-value ptime:20
last-modified-by admin@IP
last-modified-date 2015-09-15 15:09:56
sip-manipulation
  name test183
  description
  split-headers
  join-headers
  header-rule
    name manipulate183
    header-name content-type
    action manipulate
    comparison-type pattern-rule
    msg-type reply
    methods INVITE
    match-value
    new-value
  element-rule
    name check183_er

```

```

parameter-name      application/sdp
type                mime
action              sip-manip
match-val-type      any
comparison-type     pattern-rule
match-value         (.*) (?!c=IN IP4
172.16.29.44) (.*)
new-value           convert183to180
last-modified-by    admin@IP
last-modified-date  2015-09-09 14:55:43
sip-monitoring
match-any-filter    disabled
state               enabled
short-session-duration 0
monitoring-filters all
trigger-window      30
last-modified-by    admin@IP
last-modified-date  2015-09-01 11:15:07
spl-config
spl-options
last-modified-by    web_admin@IP
last-modified-date  2015-09-02 14:42:56
steering-pool
ip-address           10.64.3.163
start-port           49152
end-port             65535
realm-id             towards-sfb
network-interface
last-modified-by    admin@IP
last-modified-date  2014-11-17 13:23:58
steering-pool
ip-address           192.65.79.126
start-port           49152
end-port             65535
realm-id             SIP-Trunk
network-interface
last-modified-by    admin@IP
last-modified-date  2014-11-17 13:24:08
system-config
hostname             ACMESYSTEM
description          3820 Skype for Business IOT
location
mib-system-contact
mib-system-name
mib-system-location
snmp-enabled         enabled
enable-snmp-auth-traps disabled
enable-snmp-syslog-notify disabled
enable-snmp-monitor-traps disabled
enable-env-monitor-traps disabled
snmp-syslog-his-table-length 1
snmp-syslog-level    WARNING
system-log-level     WARNING
process-log-level    NOTICE
process-log-ip-address 0.0.0.0
process-log-port      0

```

```

collect
    sample-interval          5
    push-interval            15
    boot-state                disabled
    start-time                now
    end-time                  never
    red-collect-state         disabled
    red-max-trans             1000
    red-sync-start-time       5000
    red-sync-comp-time        1000
    push-success-trap-state   disabled
comm-monitor
    state                     disabled
    sbc-grp-id                 0
    tls-profile
    qos-enable                 enabled
    interim-qos-update         disabled
call-trace                    disabled
internal-trace                 disabled
log-filter                     all
default-gateway                 192.65.79.33
restart                          enabled
exceptions
telnet-timeout                  0
console-timeout                 0
remote-control                  enabled
cli-audit-trail                 enabled
link-redundancy-state          disabled
source-routing                  enabled
cli-more                        disabled
terminal-height                 24
debug-timeout                   0
trap-event-lifetime             0
ids-syslog-facility             -1
options
default-v6-gateway              ::
ipv6-signaling-mtu              1500
ipv4-signaling-mtu              1500
cleanup-time-of-day             00:00
snmp-engine-id-suffix
snmp-agent-mode                  v1v2
last-modified-by                 admin@IP
last-modified-date               2015-09-21 12:56:25
tls-global
    session-caching             enabled
    session-cache-timeout        12
    last-modified-by             admin@IP
    last-modified-date           2015-09-21 12:55:18
tls-profile
    name                         sfb-tls-profile
    end-entity-certificate        SBCCert1
    trusted-ca-certificates       CAcert
    cipher-list                   ALL
    verify-depth                   10
    mutual-authenticate            enabled
    tls-version                    tlsv12

```

```

options
cert-status-check                disabled
cert-status-profile-list
ignore-dead-responder            disabled
allow-self-signed-cert          disabled
last-modified-by                 admin@IP
last-modified-date               2015-09-18 16:51:10
translation-rules
id                                addplus1
type                              add
add-string                        +1
add-index                         0
delete-string                     0
delete-index                      0
last-modified-by                 admin@IP
last-modified-date               2015-03-05 12:07:49
translation-rules
id                                stripplus1
type                              delete
add-string                        0
add-index                         0
delete-string                     +1
delete-index                      0
last-modified-by                 admin@10.64.204.12
last-modified-date               2015-08-28 12:00:43
web-server-config
state                             enabled
inactivity-timeout               5
http-state                       enabled
http-port                        80
https-state                      disabled
https-port                       443
tls-profile
last-modified-by                 admin@IP
last-modified-date               2015-09-01 11:14:05
task done
ACMESYSTEM#

```

To configure TCP/RTP between SBC and Lync, two configuration elements need to be changed – the media-sec-policy should be removed from the realm-config elements and the ports of the Lync session agent needs to be modified.

```

realm-config
identifier                       towards-sfb
description
addr-prefix                      0.0.0.0
network-interfaces                slp0:0
mm-in-realm                      enabled
mm-in-network                    enabled
mm-same-ip                      enabled
mm-in-system                    enabled
bw-cac-non-mm                   disabled
msm-release                      disabled
qos-enable                      disabled
max-bandwidth                   0
fallback-bandwidth              0
max-priority-bandwidth          0

```

```

max-latency 0
max-jitter 0
max-packet-loss 0
observ-window-size 0
parent-realm
dns-realm
media-policy
media-sec-policy
srtp-msm-passthrough disabled
class-profile
in-translationid
out-translationid
in-manipulationid
out-manipulationid
average-rate-limit 0
access-control-trust-level none
invalid-signal-threshold 0
maximum-signal-threshold 0
untrusted-signal-threshold 0
nat-trust-threshold 0
max-endpoints-per-nat 0
nat-invalid-message-threshold 0
wait-time-for-invalid-register 0
deny-period 30
cac-failure-threshold 0
untrust-cac-failure-threshold 0
ext-policy-svr
diam-e2-address-realm
subscription-id-type END_USER_NONE
symmetric-latching disabled
pai-strip disabled
trunk-context
device-id
early-media-allow
enforcement-profile
additional-prefixes
restricted-latching none
restriction-mask 32
user-cac-mode none
user-cac-bandwidth 0
user-cac-sessions 0
icmp-detect-multiplier 0
icmp-advertisement-interval 0
icmp-target-ip
monthly-minutes 0
options
spl-options
accounting-enable enabled
net-management-control disabled
delay-media-update disabled
refer-call-transfer enabled
refer-notify-provisional none
dyn-refer-term disabled
codec-policy
codec-manip-in-realm disabled
codec-manip-in-network enabled

```

```

rtcp-policy
constraint-name
call-recording-server-id
session-recording-server
session-recording-required          disabled
manipulation-string
manipulation-pattern
stun-enable                          disabled
stun-server-ip                      0.0.0.0
stun-server-port                    3478
stun-changed-ip                     0.0.0.0
stun-changed-port                   3479
sip-profile
sip-isup-profile
match-media-profiles
qos-constraint
block-rtcp                          disabled
hide-egress-media-update            disabled
tcp-media-profile
monitoring-filters
node-functionality
default-location-string
alt-family-realm
pref-addr-type                      none
last-modified-by                    admin@172.18.0.119
last-modified-date                  2014-09-15 12:01:05
realm-config
  identifier                          SIP-Trunk
  description
  addr-prefix                        0.0.0.0
  network-interfaces                 s0p0:0
  mm-in-realm                        enabled
  mm-in-network                      enabled
  mm-same-ip                         enabled
  mm-in-system                       enabled
  bw-cac-non-mm                     disabled
  msm-release                        disabled
  qos-enable                         disabled
  max-bandwidth                      0
  fallback-bandwidth                0
  max-priority-bandwidth             0
  max-latency                        0
  max-jitter                         0
  max-packet-loss                   0
  observ-window-size                 0
  parent-realm
  dns-realm
  media-policy
  media-sec-policy
  srtp-msm-passthrough              disabled
  class-profile
  in-translationid
  out-translationid
  in-manipulationid
  out-manipulationid
  average-rate-limit                0

```

```

access-control-trust-level      none
invalid-signal-threshold       0
maximum-signal-threshold       0
untrusted-signal-threshold     0
nat-trust-threshold            0
max-endpoints-per-nat          0
nat-invalid-message-threshold  0
wait-time-for-invalid-register  0
deny-period                     30
cac-failure-threshold          0
untrust-cac-failure-threshold  0
ext-policy-svr
diam-e2-address-realm
subscription-id-type           END_USER_NONE
symmetric-latching             disabled
pai-strip                       disabled
trunk-context
device-id
early-media-allow
enforcement-profile
additional-prefixes
restricted-latching           none
restriction-mask               32
user-cac-mode                  none
user-cac-bandwidth             0
user-cac-sessions              0
icmp-detect-multiplier         0
icmp-advertisement-interval    0
icmp-target-ip
monthly-minutes                0
options
spl-options
accounting-enable              enabled
net-management-control         disabled
delay-media-update             disabled
refer-call-transfer            enabled
refer-notify-provisional       none
dyn-refer-term                 disabled
codec-policy
codec-manip-in-realm           disabled
codec-manip-in-network         enabled
rtcp-policy
constraint-name
call-recording-server-id
session-recording-server
session-recording-required     disabled
manipulation-string
manipulation-pattern
stun-enable                    disabled
stun-server-ip                 0.0.0.0
stun-server-port               3478
stun-changed-ip                0.0.0.0
stun-changed-port              3479
sip-profile
sip-isup-profile
match-media-profiles

```

```

qos-constraint
block-rtcp disabled
hide-egress-media-update disabled
tcp-media-profile
monitoring-filters
node-functionality
default-location-string
alt-family-realm
pref-addr-type none
last-modified-by admin@172.18.0.119
last-modified-date 2014-09-15 12:01:14
session-agent
hostname sfbmedpool.acmepacket.net
ip-address
port 5068
state enabled
app-protocol SIP
app-type
transport-method StaticTCP
realm-id towards-sfb
egress-realm-id
description
carriers
allow-next-hop-lp enabled
constraints disabled
max-sessions 0
max-inbound-sessions 0
max-outbound-sessions 0
max-burst-rate 0
max-inbound-burst-rate 0
max-outbound-burst-rate 0
max-sustain-rate 0
max-inbound-sustain-rate 0
max-outbound-sustain-rate 0
min-seizures 5
min-asr 0
time-to-resume 0
ttr-no-response 0
in-service-period 0
burst-rate-window 0
sustain-rate-window 0
req-uri-carrier-mode None
proxy-mode
redirect-action
loose-routing enabled
send-media-session enabled
response-map
ping-method OPTIONS;hops=0
ping-interval 30
ping-send-mode keep-alive
ping-all-addresses disabled
ping-in-service-response-codes
out-service-response-codes
load-balance-dns-query round-robin
options
spl-options

```



```

media-profiles
in-translationid
out-translationid
trust-me                               disabled
request-uri-headers
stop-recurse
local-response-map
ping-to-user-part
ping-from-user-part
in-manipulationid
out-manipulationid
manipulation-string
manipulation-pattern
p-asserted-id
trunk-group
max-register-sustain-rate              0
early-media-allow
invalidate-registrations                disabled
rfc2833-mode                            none
rfc2833-payload                          0
codec-policy
enforcement-profile
refer-call-transfer                     enabled
refer-notify-provisional                none
reuse-connections                       NONE
tcp-keepalive                           none
tcp-reconn-interval                     0
max-register-burst-rate                  0
register-burst-window                    0
sip-profile
sip-isup-profile
kpml-interworking                       inherit
monitoring-filters
session-recording-server
session-recording-required              disabled
last-modified-by                        admin@172.18.0.142

```

IPv4 with TCP/RTP

The following section provides information on configuration required on the SBC to route TCP/RTP based calls to and from the Skype for Business environment.

```

certificate-record
  name                               CAcert
  organization                         Tekvizion
  common-name                          sfblabdm-DC-CA-1
  key-size                              2048
certificate-record
  name                               SBCcert1
  organization                         Tekvizion
  common-name                          oracle.sfblabdm.local
  key-size                              2048
  alternate-name                       oraclepool.sfblabdm.local
codec-policy
  name                                 For488

```

```

    allow-codecs G729 PCMU:no telephone-event
PCMA:no
    order-codecs G729 telephone-event
codec-policy
    name TC
    allow-codecs PCMA telephone-event PCMU:no
    add-codecs-on-egress PCMA telephone-event G729:no
    order-codecs PCMA telephone-event G729:no
codec-policy
    name TCPCMU
    allow-codecs PCMU telephone-event PCMA:no
    order-codecs PCMU telephone-event
filter-config
    name all
    user *
local-policy
    from-address *
    to-address *
    source-realm SIP-Trunk
    policy-attribute
        next-hop Medpool.sfbblabdm.local
        realm towards-sfb
        action replace-uri
        app-protocol SIP
local-policy
    from-address *
    to-address med1.sfbblabdm.local
    source-realm SIP-Trunk
    description For Referred-Party Header
    policy-attribute
        next-hop med1.sfbblabdm.local
        realm towards-sfb
        action replace-uri
        app-protocol SIP
local-policy
    from-address *
    to-address med2.sfbblabdm.local
    source-realm SIP-Trunk
    policy-attribute
        next-hop med2.sfbblabdm.local
        realm towards-sfb
        action replace-uri
local-policy
    from-address *
    to-address *
    source-realm towards-sfb
    policy-attribute
        next-hop 63.87.147.48
        realm SIP-Trunk
        action replace-uri
        app-protocol SIP
media-manager
media-sec-policy
    name rtp
media-sec-policy
    name sfb-srtp
    inbound

```

```

        profile          sdes-profile
        mode             srtp
        protocol         sdes
    outbound
        profile          sdes-profile
        mode             srtp
        protocol         sdes
network-interface
    name                s0p0
    description         For SIP-Trunk
    ip-address          192.65.79.126
    netmask             255.255.255.224
    gateway             192.65.79.97
    hip-ip-list         192.65.79.126
    icmp-address        192.65.79.126
network-interface
    name                s1p0
    description         Facing Skype for Business
    hostname            oracle.sfbldm.local
    ip-address          10.64.3.163
    netmask             255.255.0.0
    gateway             10.64.1.1
    dns-ip-primary     172.16.29.42
    dns-domain          sfbldm.local
    hip-ip-list         10.64.3.163
    icmp-address        10.64.3.163
    ssh-address         10.64.3.163
network-interface
    name                wancom1
    pri-utility-addr    169.254.1.1
    sec-utility-addr    169.254.1.2
    netmask             255.255.255.252
network-interface
    name                wancom2
    pri-utility-addr    169.254.2.1
    sec-utility-addr    169.254.2.2
    netmask             255.255.255.252
phy-interface
    name                s0p0
    operation-type      Media
    virtual-mac         00:08:25:04:0d:1e
phy-interface
    name                s1p0
    operation-type      Media
    slot                1
    virtual-mac         00:08:25:04:0d:1f
phy-interface
    name                wancom1
    port                1
    duplex-mode         speed
    wancom-health-score 8
phy-interface
    name                wancom2
    port                2
    duplex-mode         speed

```

```

wancom-health-score          9
playback-config
  name                       transferrbt
  entry
    encoding                  PCMU
    filename                   US_ringbackPCMU.raw
realm-config
  identifier                  SIP-Trunk
  network-interfaces          s0p0:0
  mm-in-realm                 enabled
  restricted-latching         sdp
realm-config
  identifier                  towards-sfb
  network-interfaces          slp0:0
  mm-in-realm                 enabled
  refer-call-transfer         enabled
redundancy-config
  peer
    name                      acme_ap3820
    type                      Primary
    destination
      address                  169.254.1.1:9090
      network-interface        wancom1:0
    destination
      address                  169.254.2.1:9090
      network-interface        wancom2:0
  peer
    name                      acme_ap3820_b
    type                      Secondary
    destination
      address                  169.254.1.2:9090
      network-interface        wancom1:0
    destination
      address                  169.254.2.2:9090
      network-interface        wancom2:0
response-map
  name                       Changefor408324
  entries
    rcv-code                   503
    xmit-code                   404
    reason                      Not Found
response-map
  name                       change183to180
  entries
    rcv-code                   183
    xmit-code                   180
    reason                      Ringing
sdes-profile
  name                       sdes-profile
session-agent
  hostname                    63.87.147.48
  ip-address                   63.87.147.48
  port                        5071
  realm-id                    SIP-Trunk
  ping-method                  OPTIONS
  ping-interval                30
  out-translationid            stripplus1

```

```

session-agent
  hostname                               Medpool.sfbblabdm.local
  transport-method                        StaticTCP
  realm-id                                towards-sfb
  ping-method                             OPTIONS
  ping-interval                           60
  ping-all-addresses                     enabled
  out-translationid                       addplus1
  stop-recurse                            480
  refer-call-transfer                     enabled
  tcp-keepalive                           disabled
session-translation
  id                                       addplus1
  rules-calling                           addplus1
  rules-called                             addplus1
session-translation
  id                                       stripplus1
  rules-calling                           stripplus1
  rules-called                             stripplus1
sip-config
  home-realm-id                           towards-sfb
  registrar-domain                        *
  registrar-host                           *
  options                                  max-udp-length=0
sip-feature
  name                                     100rel
  realm                                    SIP-Trunk
  require-mode-inbound                    Pass
  require-mode-outbound                   Pass
sip-interface
  realm-id                                 SIP-Trunk
  sip-port
    address                               192.65.79.126
  out-manipulationid                      NATandChangeContact
sip-interface
  realm-id                                 towards-sfb
  sip-port
    address                               10.64.3.163
    transport-protocol                    TCP
  options                                  100rel-interworking
  in-manipulationid                       ForEarlyMedia
  out-manipulationid                       outManipToSFB
  response-map                             Changefor408324
sip-manipulation
  name                                     Addpcmato183
  header-rule
    name                                   check1for83
    header-name                            @status-line
    action                                  manipulate
    comparison-type                        pattern-rule
    element-rule
      name                                  isit183
      type                                   status-code
      action                                  sip-manip
      comparison-type                        pattern-rule
      match-value                           183
      new-value                              fix183sdp

```

```

sip-manipulation
  name ChangeContact
  header-rule
    name StoreFromnumber
    header-name From
    action manipulate
    element-rule
      name StoreFromnumber_er
      type uri-user-only
      action store
  header-rule
    name ChangeContact
    header-name Contact
    action manipulate
    element-rule
      name ChangeContact_er
      type uri-user
      action add
      new-value
$StoreFromnumber.$StoreFromnumber_er.$0
sip-manipulation
  name ChangeRecvOnlyToInactive
  header-rule
    name changeSDP
    header-name Content-Type
    action manipulate
    msg-type reply
    methods INVITE
    element-rule
      name RecvOnlyToInactive
      parameter-name application/sdp
      type mime
      action find-replace-all
      comparison-type pattern-rule
      match-value a=recvonly
      new-value a=inactive
sip-manipulation
  name Changeinactosendonly
  description Change inactive to sendonly for
pstn tran
  header-rule
    name changeSDP
    header-name Content-Type
    action manipulate
    msg-type request
    methods INVITE
    element-rule
      name inacttosendonly
      parameter-name application/sdp
      type mime
      action find-replace-all
      comparison-type pattern-rule
      match-value a=inactive
      new-value a=sendonly
sip-manipulation
  name Check183
  header-rule

```

```

name check183
header-name @status-line
action manipulate
comparison-type pattern-rule
element-rule
    name is183
    type status-code
    action sip-manip
    comparison-type pattern-rule
    match-value 183
    new-value Modcline

sip-manipulation
    name ForEarlyMedia
    header-rule
        name delsupported
        header-name Supported
        action delete
        msg-type request
        methods INVITE
    header-rule
        name addrequireinINVITE
        header-name Require
        action add
        msg-type request
        methods INVITE
        new-value 100rel
    header-rule
        name Options
        header-name From
        action sip-manip
        new-value OptionsResponseLocally
    header-rule
        name Change183
        header-name From
        action sip-manip
        new-value Check183

sip-manipulation
    name Map183
    header-rule
        name check183
        header-name @status-line
        action store
        comparison-type pattern-rule
        element-rule
            name is183
            type status-code
            action replace
            comparison-type pattern-rule
            match-value 183
            new-value 180

sip-manipulation
    name Modcline
    mime-sdp-rule
        name test
        action manipulate
        comparison-type pattern-rule
        sdp-media-rule

```



```

new-value Addpcmato183
sip-manipulation
  name
  header-rule
    name
    header-name
    action
    msg-type
    element-rule
      name
      type
      action
      new-value
    header-rule
      name
      header-name
      action
      msg-type
      element-rule
        name
        type
        action
        new-value
sip-manipulation
  name
  header-rule
    name
    header-name
    action
    msg-type
    methods
    new-value
sip-manipulation
  name
  header-rule
    name
    header-name
    action
    msg-type
    methods
  header-rule
    name
    header-name
    action
    comparison-type
    msg-type
    methods
    match-value
    new-value
<sip:5712935325+@+${LOCAL_IP}>
sip-manipulation
  name
  header-rule
    name
    header-name
    action
    msg-type

```

```

        methods INVITE
    header-rule
        name addPAIifHistoryInfo
        header-name P-Asserted-Identity
        action add
        comparison-type pattern-rule
        msg-type request
        methods INVITE
        match-value $checkForHistoryInfo
        new-value
<sip:5712935325+@+$LOCAL_IP+>
sip-manipulation
    name convert183to180
    header-rule
        name delSDP
        header-name Content-Type
        action manipulate
        comparison-type case-insensitive
        element-rule
            name del183SDP
            parameter-name application/sdp
            type mime
            action delete-element
    header-rule
        name delContentType
        header-name Content-Type
        action manipulate
        element-rule
            name delCT
            parameter-name *
            type header-param
            action delete-header
    header-rule
        name change487to486
        header-name @status-line
        action manipulate
        element-rule
            name modStatusCode
            type status-code
            action replace
            match-value 183
            new-value 180
        element-rule
            name modReasonPhrase
            type reason-phrase
            action replace
            comparison-type case-insensitive
            match-value Session Progress
            new-value Ringing
sip-manipulation
    name fix183sdp
    description To bypass the reinvoke
    header-rule
        name Checkforpcma
        header-name Content-type
        action store
        element-rule

```

```

name Checkpcmaexists
parameter-name application/sdp
type mime
action store
match-value (a=rtpmap:8
PCMA/8000)
header-rule
name Addpcma
header-name Content-Type
action manipulate
comparison-type boolean
match-
value !$Checkforpcma.$Checkpcmaexists
element-rule
name Addpcma
parameter-name application/sdp
type mime
action find-replace-all
comparison-type pattern-rule
match-value a=rtpmap:0
PCMU/8000
new-value "a=rtpmap:0
PCMU/8000"+$CRLF+"a=rtpmap:8 PCMA/8000"
mime-sdp-rule
name modmline
action manipulate
sdp-media-rule
name modmline_m
media-type audio
action manipulate
sdp-line-rule
name
change_payload
type m
action find-
replace-all
comparison-type pattern-
rule
match-value
new-value audio+$2+"
RTP/AVP 0 8 101"
sip-manipulation
name modcontact
header-rule
name modcontact_er
header-name Contact
action manipulate
comparison-type pattern-rule
element-rule
name moder
type uri-user
action replace
comparison-type pattern-rule
match-value ((\+1) (\+1\d+))
new-value $3
element-rule

```

```

        name                modanon
        type                uri-user
        action              replace
        comparison-type    pattern-rule
        match-value        \+1anonymous
        new-value           anonymous
sip-manipulation
  name                    outManipToSFB
  header-rule
    name                  From
    header-name           From
    action                manipulate
    msg-type              request
    element-rule
      name                From_header
      type                uri-host
      action              replace
      new-value
oracle.sfbldm.local
  element-rule
    name                  modanonfrom
    type                uri-user
    action              replace
    comparison-type    pattern-rule
    match-value        \+1anonymous
    new-value           anonymous
  header-rule
    name                  To
    header-name           To
    action                manipulate
    msg-type              request
    element-rule
      name                remplusone
      type                uri-user
      action              replace
      comparison-type    pattern-rule
      match-value        \+1$
    element-rule
      name                To
      type                uri-host
      action              replace
      new-value
medpool.sfbldm.local
  header-rule
    name                  fixcontact
    header-name           Contact
    action                manipulate
    element-rule
      name                updatecontact
      type                uri-host
      action              replace
      new-value
oracle.sfbldm.local
  header-rule
    name                  fixcontactuser
    header-name           Contact
    action                manipulate

```

comparison-type	pattern-rule
msg-type	request
methods	INVITE
element-rule	
name	user
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	(.*)
new-value	"*1"+\$ORIGINAL
element-rule	
name	moder
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	((\+1) (\+1\d+))
new-value	\$3
element-rule	
name	modanoncon
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	\+1anonymous
new-value	anonymous
header-rule	
name	Ruri
header-name	Request-URI
action	manipulate
msg-type	request
element-rule	
name	remplusone
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	\+1\$
header-rule	
name	addReplaces
header-name	Supported
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
new-value	\$ORIGINAL+, replaces
header-rule	
name	addUpdate
header-name	Allow
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
new-value	\$ORIGINAL+, UPDATE
header-rule	
name	addSessionExpires
header-name	Session-Expires
action	add
msg-type	request
methods	INVITE

```

        new-value 300
    header-rule
        name addingptime
        header-name From
        action sip-manip
        msg-type request
        new-value replacemptimewithptime
sip-manipulation
    name replacemptimewithptime
    mime-sdp-rule
        name addptime
        msg-type request
        action manipulate
        sdp-media-rule
            name addSDPptime
            media-type audio
            action manipulate
            sdp-line-rule
                name addptimelr
                type a
                action find-
replace-all
    match-value
maxptime:20
    new-value ptime:20
sip-manipulation
    name test183
    header-rule
        name manipulate183
        header-name content-type
        action manipulate
        comparison-type pattern-rule
        msg-type reply
        methods INVITE
        element-rule
            name check183_er
            parameter-name application/sdp
            type mime
            action sip-manip
            comparison-type pattern-rule
            match-value (.*) (?!c=IN IP4
172.16.29.44) (.*)
    new-value convert183to180
sip-monitoring
    monitoring-filters all
spl-config
steering-pool
    ip-address 10.64.3.163
    start-port 49152
    end-port 65535
    realm-id towards-sfb
steering-pool
    ip-address 192.65.79.126
    start-port 49152
    end-port 65535
    realm-id SIP-Trunk
system-config

```

```

hostname                ACMESYSTEM
description              3820 Skype for Business IOT
process-log-level       DEBUG
default-gateway         192.65.79.33
source-routing          enabled
tls-profile
  name                   sfb-tls-profile
  end-entity-certificate SBCcert1
  trusted-ca-certificates CAcert
  mutual-authenticate   enabled
  tls-version            tlsv12
translation-rules
  id                     addplus1
  type                   add
  add-string             +1
translation-rules
  id                     stripplus1
  type                   delete
  delete-string          +1

```

IPv6 based signaling for communication between Skype for Business and Oracle ESBC

IPv6 with TLS/SRTP

***** Note: SBC requires IPv6-IPv4 interworking license in order for this to work**

```
certificate-record
  name          CAcert
  organization   Tekvizion
  common-name    sflabdm-DC-CA-1
  key-size       2048
certificate-record
  name          CAcert2
  organization   Tekvizion
  common-name    sflabdm-DC-CA-1
  key-size       2048
certificate-record
  name          CAcertpool
  organization   Tekvizion
  common-name    sflabdm-DC-CA-1
  key-size       2048
certificate-record
  name          SBCcert1
  organization   Tekvizion
  common-name    oracle.sflabdm.local
  key-size       2048
  alternate-name oraclepool.sflabdm.local
certificate-record
  name          SBCcert2
  organization   Tekvizion
  common-name    oracle.sflabdm.local
  key-size       2048
  alternate-name DNS:oraclepool.sflabdm.local
codec-policy
  name          For488
  allow-codecs  G729 PCMU:no telephone-event PCMA:no
  order-codecs G729 telephone-event
codec-policy
  name          TC
  allow-codecs  PCMA telephone-event PCMU:no
  add-codecs-on-egress PCMA telephone-event G729:no
  order-codecs PCMA telephone-event G729:no
codec-policy
  name          TCPCMU
  allow-codecs  PCMU telephone-event PCMA:no
  order-codecs PCMU telephone-event
filter-config
  name          all
  user          *
filter-config
  name          ip6
  address       ::
  user          *
local-policy
  from-address  *
  to-address    *
  source-realm  SIP-Trunk
  policy-attribute
    next-hop    Medpool.sflabdm.local
    realm       towards-sfb
    action       replace-uri
    app-protocol SIP
local-policy
```



```

    from-address      *
    to-address        med1.sfbblabdm.local
    source-realm      SIP-Trunk
    description        For Referred-Party Header
    policy-attribute
        next-hop          med1.sfbblabdm.local
        realm              towards-sfb
        action              replace-uri
        app-protocol       SIP
local-policy
    from-address      *
    to-address        med2.sfbblabdm.local
    source-realm      SIP-Trunk
    policy-attribute
        next-hop          med2.sfbblabdm.local
        realm              towards-sfb
        action              replace-uri
local-policy
    from-address      *
    to-address        *
    source-realm      towards-sfb
    policy-attribute
        next-hop          63.87.147.48
        realm              SIP-Trunk
        action              replace-uri
        app-protocol       SIP
media-manager
media-sec-policy
    name              rtp
media-sec-policy
    name              sfb-srtp
    inbound
        profile          sdes-profile
        mode              srtp
        protocol          sdes
    outbound
        profile          sdes-profile
        mode              srtp
        protocol          sdes
network-interface
    name              s0p0
    description        For SIP-Trunk
    ip-address         192.65.79.126
    netmask            255.255.255.224
    gateway            192.65.79.97
    hip-ip-list        192.65.79.126
    icmp-address       192.65.79.126
network-interface
    name              s1p0
    description        Facing Skype for Business
    hostname           oracle.sfbblabdm.local
    ip-address         2620:96:c000:1::10/64
    netmask            ffff:fff:fff:fff::
    gateway            2620:96:c000:1::1
    dns-ip-primary     2620:96:c000:8:e840:68d3:2a77:3d2d
    dns-domain         sfbblabdm.local
    hip-ip-list        2620:96:c000:1::10
    icmp-address       2620:96:c000:1::10
    ssh-address        10.64.3.163
network-interface
    name              wancom1
    pri-utility-addr   169.254.1.1
    sec-utility-addr   169.254.1.2
    netmask            255.255.255.252

```

```

network-interface
  name wancom2
  pri-utility-addr 169.254.2.1
  sec-utility-addr 169.254.2.2
  netmask 255.255.255.252
phy-interface
  name s0p0
  operation-type Media
  virtual-mac 00:08:25:04:0d:1e
phy-interface
  name slp0
  operation-type Media
  slot 1
  virtual-mac 00:08:25:04:0d:1f
phy-interface
  name wancom1
  port 1
  duplex-mode
  speed
  wancom-health-score 8
phy-interface
  name wancom2
  port 2
  duplex-mode
  speed
  wancom-health-score 9
playback-config
  name transferrbt
  entry
    encoding PCMU
    filename US_ringbackPCMU.raw
realm-config
  identifier SIP-Trunk
  network-interfaces s0p0:0
  mm-in-realm enabled
  media-sec-policy rtp
realm-config
  identifier towards-sfb
  addr-prefix ::
  network-interfaces slp0:0
  mm-in-realm enabled
  media-sec-policy sfb-srtp
  refer-call-transfer enabled
redundancy-config
  peer
    name Acmesystem1
    type Primary
    destination
      address 169.254.1.1:9090
      network-interface wancom1:0
    destination
      address 169.254.2.1:9090
      network-interface wancom2:0
  peer
    name acmesystem3
    type Secondary
    destination
      address 169.254.1.2:9090
      network-interface wancom1:0
    destination
      address 169.254.2.2:9090
      network-interface wancom2:0
response-map
  name change183to180

```

```

entries
    recv-code          183
    xmit-code          180
    reason              Ringing
sdes-profile
    name                sdes-profile
session-agent
    hostname            63.87.147.48
    ip-address          63.87.147.48
    port                5071
    realm-id            SIP-Trunk
    ping-method         OPTIONS
    ping-interval       30
    out-translationid   stripplus1
session-agent
    hostname            Medpool.sfb labdm.local
    port                5067
    transport-method    StaticTLS
    realm-id            towards-sfb
    ping-method         OPTIONS
    ping-interval       60
    out-translationid   addplus1
    stop-recurse        480
    refer-call-transfer enabled
    tcp-keepalive       disabled
session-timer-profile
    name                towards-sfb
    session-expires     900
    request-refresher   none
session-translation
    id                  addplus1
    rules-calling        addplus1
    rules-called         addplus1
session-translation
    id                  stripplus1
    rules-calling        stripplus1
    rules-called         stripplus1
sip-config
    home-realm-id       towards-sfb
    registrar-domain    *
    registrar-host      *
    options              max-udp-length=0
sip-feature
    name                100rel
    realm                SIP-Trunk
    require-mode-inbound Pass
    require-mode-outbound Pass
sip-interface
    realm-id            SIP-Trunk
    sip-port
        address          192.65.79.126
    out-manipulationid NATandChangeContact
sip-interface
    realm-id            towards-sfb
    sip-port
        address          2620:96:c000:1::10
        port              5067
        transport-protocol TLS
        tls-profile        sfb-tls-profile
    options              100rel-interworking
    in-manipulationid   ForEarlyMedia
    out-manipulationid outManipToSFB
    session-timer-profile towards-sfb
sip-manipulation

```

```

name Addpcmato183
header-rule
  name
  header-name
  action
  comparison-type
  element-rule
    name
    type
    action
    comparison-type
    match-value
    new-value
sip-manipulation
  name
  header-rule
    name
    header-name
    action
    element-rule
      name
      type
      action
  header-rule
    name
    header-name
    action
    element-rule
      name
      type
      action
      new-value
$StoreFromnumber.$StoreFromnumber_er.$0
sip-manipulation
  name
  header-rule
    name
    header-name
    action
    msg-type
    methods
    element-rule
      name
      parameter-name
      type
      action
      comparison-type
      match-value
      new-value
sip-manipulation
  name
  description
  header-rule
    name
    header-name
    action
    msg-type
    methods
    element-rule
      name
      parameter-name
      type
      action
      comparison-type

```

```

Addpcmato183
check1for83
@status-line
manipulate
pattern-rule
isit183
status-code
sip-manip
pattern-rule
183
fix183sdp
ChangeContact
StoreFromnumber
From
manipulate
StoreFromnumber_er
uri-user-only
store
ChangeContact
Contact
manipulate
ChangeContact_er
uri-user
add
ChangeRecvOnlyToInactive
changeSDP
Content-Type
manipulate
reply
INVITE
RecvOnlyToInactive
application/sdp
mime
find-replace-all
pattern-rule
a=recvonly
a=inactive
Changeinactosendonly
Change inactive to sendonly for pstn tran
changeSDP
Content-Type
manipulate
request
INVITE
inacttosendonly
application/sdp
mime
find-replace-all
pattern-rule

```

```

                                match-value
                                new-value
                                a=inactive
                                a=sendonly
sip-manipulation
  name
  header-rule
    name
    header-name
    action
    comparison-type
    element-rule
      name
      type
      action
      comparison-type
      match-value
      new-value
    name
    type
    action
    comparison-type
    match-value
    new-value
                                Check183
                                check183
                                @status-line
                                manipulate
                                pattern-rule
                                is183
                                status-code
                                sip-manip
                                pattern-rule
                                183
                                Modcline

sip-manipulation
  name
  header-rule
    name
    header-name
    action
    msg-type
    methods
  header-rule
    name
    header-name
    action
    msg-type
    methods
    new-value
  header-rule
    name
    header-name
    action
    new-value
  header-rule
    name
    header-name
    action
    new-value
                                ForEarlyMedia
                                delsupported
                                Supported
                                delete
                                request
                                INVITE
                                addressrequireinINVITE
                                Require
                                add
                                request
                                INVITE
                                100rel
                                Options
                                From
                                sip-manip
                                OptionsResponseLocally
                                Change183
                                From
                                sip-manip
                                Check183

sip-manipulation
  name
  header-rule
    name
    header-name
    action
    comparison-type
    element-rule
      name
      type
      action
      comparison-type
      match-value
      new-value
    name
    type
    action
    comparison-type
    match-value
    new-value
                                Map183
                                check183
                                @status-line
                                store
                                pattern-rule
                                is183
                                status-code
                                replace
                                pattern-rule
                                183
                                180

sip-manipulation
  name
  mime-sdp-rule
    name
    action
    comparison-type
    sdp-media-rule
      name
      media-type
                                Modcline
                                test
                                manipulate
                                pattern-rule
                                checkcline
                                audio

```

```

        action
        comparison-type
        sdp-line-rule
            name
            type
            action
            comparison-type
            match-value
^((?! (2620:0096:c000:1::10|2620:96:c000:8:24b6:215d:7023:ec7d|2620:96:c000:8:55d6:58f5:b47:fe82))) *$
        new-value
            header-rule
                name
                header-name
                action
                match-value
                new-value
            manipulate
            pattern-rule
                detectc
                c
                sip-manip
                pattern-rule
            convert183to180
            checkcl
            Content-Length
            sip-manip
            0
            convert183to180
sip-manipulation
    name
    description
    header-rule
        name
        header-name
        action
        new-value
    header-rule
        name
        header-name
        action
        msg-type
        methods
    header-rule
        name
        header-name
        action
        msg-type
        methods
        new-value
    header-rule
        name
        header-name
        action
        msg-type
        methods
        new-value
    header-rule
        name
        header-name
        action
        msg-type
        methods
        new-value
    header-rule
        name
        header-name
        action
        msg-type
        element-rule
            name
            NATandChangeContact
            NAT plus Change Contact
            doNATfortrunk
            From
            sip-manip
            NATting
            delsupported
            Supported
            delete
            request
            INVITE
            addPAIonCallForwards
            From
            sip-manip
            request
            INVITE
            addPAIonCallForwards
            addPAIRefers
            From
            sip-manip
            request
            INVITE
            addPAIRefers
            ChangeContactHeader
            From
            sip-manip
            ChangeContact
            modify183sdp
            From
            sip-manip
            Addpcmato183
    NATting
    From
    From
    manipulate
    request
    From_header

```

```

                type                uri-host
                action               replace
                new-value            $LOCAL_IP
header-rule
    name
header-name    To
action         manipulate
msg-type      request
element-rule
    name
    type
    action     uri-host
    new-value  replace
                $REMOTE_IP
sip-manipulation
    name                OptionsResponseLocally
header-rule
    name                rejectOptions
header-name           request-uri
action                reject
msg-type              request
methods               OPTIONS
new-value             200: OK
sip-manipulation
    name                addPAIRefers
header-rule
    name                checkReferredBy
header-name           REFERRED-BY
action                store
msg-type              request
methods               INVITE
header-rule
    name                addPAIifReferredBy
header-name           P-Asserted-Identity
action                add
comparison-type       pattern-rule
msg-type              request
methods               INVITE
match-value           $checkReferredBy
new-value             <sip:5712935325+@+$LOCAL_IP+>
sip-manipulation
    name                addPAIonCallForwards
header-rule
    name                checkForHistoryInfo
header-name           History-Info
action                store
msg-type              request
methods               INVITE
header-rule
    name                addPAIifHistoryInfo
header-name           P-Asserted-Identity
action                add
comparison-type       pattern-rule
msg-type              request
methods               INVITE
match-value           $checkForHistoryInfo
new-value             <sip:5712935325+@+$LOCAL_IP+>
sip-manipulation
    name                changecline
mime-sdp-rule
    name                cline
msg-type              reply
action                manipulate
comparison-type       pattern-rule
sdp-session-rule

```

```

name con
action manipulate
comparison-type pattern-rule
sdp-line-rule
    name conl
    type c
    action find-replace-all
    comparison-type pattern-rule
    match-value (IN IP6
anonymous.invalid)
    new-value "IN IP6 ::"
sip-manipulation
    name convert183to180
    header-rule
        name delSDP
        header-name Content-Type
        action manipulate
        comparison-type case-insensitive
        element-rule
            name del183SDP
            parameter-name application/sdp
            type mime
            action delete-element
    header-rule
        name delContentType
        header-name Content-Type
        action manipulate
        element-rule
            name delCT
            parameter-name *
            type header-param
            action delete-header
    header-rule
        name change487to486
        header-name @status-line
        action manipulate
        element-rule
            name modStatusCode
            type status-code
            action replace
            match-value 183
            new-value 180
        element-rule
            name modReasonPhrase
            type reason-phrase
            action replace
            comparison-type case-insensitive
            match-value Session Progress
            new-value Ringing
sip-manipulation
    name fix183sdp
    description To bypass the reinvoke
    header-rule
        name Checkforpcma
        header-name Content-type
        action store
        element-rule
            name Checkpcmaexists
            parameter-name application/sdp
            type mime
            action store
            match-value (a=rtpmap:8 PCMA/8000)
    header-rule
        name Addpcma

```



```

header-name Content-Type
action manipulate
comparison-type boolean
match-value !$Checkforpcma.$Checkpcmaexists
element-rule
  name Addpcma
  parameter-name application/sdp
  type mime
  action find-replace-all
  comparison-type pattern-rule
  match-value a=rtpmap:0 PCMU/8000
  new-value "a=rtpmap:0
PCMU/8000"+$CRLF+"a=rtpmap:8 PCMA/8000"
mime-sdp-rule
  name modmline
  action manipulate
  sdp-media-rule
    name modmline_m
    media-type audio
    action manipulate
    sdp-line-rule
      name change_payload
      type m
      action find-replace-all
      comparison-type pattern-rule
      match-value ^(audio) ( [0-
9]{4,5}) ( RTP/AVP 0 101)$
      new-value audio+$2+" RTP/AVP
0 8 101"
sip-manipulation
  name modcontact
  header-rule
    name modcontact_er
    header-name Contact
    action manipulate
    comparison-type pattern-rule
    element-rule
      name moder
      type uri-user
      action replace
      comparison-type pattern-rule
      match-value ((\+1) (\+1\d+))
      new-value $3
    element-rule
      name modanon
      type uri-user
      action replace
      comparison-type pattern-rule
      match-value \+1anonymous
      new-value anonymous
sip-manipulation
  name outManipToSFB
  header-rule
    name From
    header-name From
    action manipulate
    msg-type request
    element-rule
      name From_header
      type uri-host
      action replace
      new-value oracle.sfblabdm.local
    element-rule
      name modanonfrom

```

```

        type uri-user
        action replace
        comparison-type pattern-rule
        match-value \+lanonymous
        new-value anonymous
header-rule
    name To
    header-name To
    action manipulate
    msg-type request
    element-rule
        name remplusone
        type uri-user
        action replace
        comparison-type pattern-rule
        match-value \+1$
    element-rule
        name To
        type uri-host
        action replace
        new-value medpool.sfbldm.local
header-rule
    name fixcontact
    header-name Contact
    action manipulate
    element-rule
        name updatecontact
        type uri-host
        action replace
        new-value oracle.sfbldm.local
header-rule
    name fixcontactuser
    header-name Contact
    action manipulate
    comparison-type pattern-rule
    msg-type request
    methods INVITE
    element-rule
        name user
        type uri-user
        action replace
        comparison-type pattern-rule
        match-value (.*)
        new-value "+1"+$ORIGINAL
    element-rule
        name moder
        type uri-user
        action replace
        comparison-type pattern-rule
        match-value ((\+1)(\+1\d+))
        new-value $3
    element-rule
        name modanoncon
        type uri-user
        action replace
        comparison-type pattern-rule
        match-value \+lanonymous
        new-value anonymous
header-rule
    name Ruri
    header-name Request-URI
    action manipulate
    msg-type request
    element-rule

```

```

        name                remplusone
        type                uri-user
        action              replace
        comparison-type     pattern-rule
        match-value         \+1$

header-rule
  name                    addReplaces
  header-name             Supported
  action                  manipulate
  comparison-type         pattern-rule
  msg-type                request
  methods                 INVITE
  new-value               $ORIGINAL+, replaces

header-rule
  name                    addUpdate
  header-name             Allow
  action                  manipulate
  comparison-type         pattern-rule
  msg-type                request
  methods                 INVITE
  new-value               $ORIGINAL+, UPDATE

header-rule
  name                    addingptime
  header-name             From
  action                  sip-manip
  msg-type                request
  new-value               replacemptimewithptime

header-rule
  name                    forconsul
  header-name             From
  action                  sip-manip
  msg-type                reply
  new-value               changecline

sip-manipulation
  name                    replacemptimewithptime
  mime-sdp-rule
    name                  addptime
    msg-type              request
    action                manipulate
    sdp-media-rule
      name                addSDPptime
      media-type          audio
      action              manipulate
      sdp-line-rule
        name              addptimelr
        type              a
        action            find-replace-all
        match-value       maxptime:20
        new-value         ptime:20

sip-manipulation
  name                    test183
  header-rule
    name                  manipulate183
    header-name           content-type
    action                manipulate
    comparison-type       pattern-rule
    msg-type              reply
    methods               INVITE
    element-rule
      name                check183_er
      parameter-name      application/sdp
      type                mime
      action              sip-manip
      comparison-type     pattern-rule

```

```

172.16.29.44) (.*)          match-value          (.*) (?!c=IN IP4
                             new-value          convert183to180
sip-monitoring
  monitoring-filters        all,ip6
spl-config
steering-pool
  ip-address                192.65.79.126
  start-port                49152
  end-port                  65535
  realm-id                  SIP-Trunk
steering-pool
  ip-address                2620:96:c000:1::10
  start-port                49152
  end-port                  65535
  realm-id                  towards-sfb
system-config
  hostname                  ACMESYSTEM
  description                3820 Skype for Business IOT
  default-gateway            192.65.79.33
  source-routing             enabled
  default-v6-gateway         2620:96:c000:1::1
  ipv6-signaling-mtu        1476
tls-global
  session-caching            enabled
tls-profile
  name                      sfb-tls-profile
  end-entity-certificate     SBCcert1
  trusted-ca-certificates    CAcert
  mutual-authenticate        enabled
  tls-version                tlsv12
translation-rules
  id                        addplus1
  type                      add
  add-string                 +1
translation-rules
  id                        stripplus1
  type                      delete
  delete-string              +1

```

IPv6 with TCP/RTP

```

certificate-record
  name                      CAcert
  organization                Tekvizion
  common-name                 sflabdm-DC-CA-1
  key-size                    2048
certificate-record
  name                      CAcert2
  organization                Tekvizion
  common-name                 sflabdm-DC-CA-1
  key-size                    2048
certificate-record
  name                      CAcertpool
  organization                Tekvizion
  common-name                 sflabdm-DC-CA-1
  key-size                    2048
certificate-record
  name                      SBCcert1
  organization                Tekvizion
  common-name                 oracle.sflabdm.local
  key-size                    2048
  alternate-name              oraclepool.sflabdm.local

```

```

certificate-record
  name SBCcert2
  organization Tekvizion
  common-name oracle.sfbblabdm.local
  key-size 2048
  alternate-name DNS:oraclepool.sfbblabdm.local
codec-policy
  name For488
  allow-codecs G729 PCMU:no telephone-event PCMA:no
  order-codecs G729 telephone-event
codec-policy
  name TC
  allow-codecs PCMA telephone-event PCMU:no
  add-codecs-on-egress PCMA telephone-event G729:no
  order-codecs PCMA telephone-event G729:no
codec-policy
  name TCPCMU
  allow-codecs PCMU telephone-event PCMA:no
  order-codecs PCMU telephone-event
filter-config
  name all
  user *
filter-config
  name ip6
  address ::
  user *
local-policy
  from-address *
  to-address *
  source-realm SIP-Trunk
  policy-attribute
    next-hop Medpool.sfbblabdm.local
    realm towards-sfb
    action replace-uri
    app-protocol SIP
local-policy
  from-address *
  to-address med1.sfbblabdm.local
  source-realm SIP-Trunk
  description For Referred-Party Header
  policy-attribute
    next-hop med1.sfbblabdm.local
    realm towards-sfb
    action replace-uri
    app-protocol SIP
local-policy
  from-address *
  to-address med2.sfbblabdm.local
  source-realm SIP-Trunk
  policy-attribute
    next-hop med2.sfbblabdm.local
    realm towards-sfb
    action replace-uri
local-policy
  from-address *
  to-address *
  source-realm towards-sfb
  policy-attribute
    next-hop 63.87.147.48
    realm SIP-Trunk
    action replace-uri
    app-protocol SIP
media-manager
media-sec-policy

```

```

name rtp
media-sec-policy
  name sfb-srtp
  inbound
    profile sdes-profile
    mode any
    protocol sdes
  outbound
    profile sdes-profile
    mode any
    protocol sdes
network-interface
  name s0p0
  description For SIP-Trunk
  ip-address 192.65.79.126
  netmask 255.255.255.224
  gateway 192.65.79.97
  hip-ip-list 192.65.79.126
  icmp-address 192.65.79.126
network-interface
  name s1p0
  description Facing Skype for Business
  hostname oracle.sfblabdm.local
  ip-address 2620:96:c000:1::10/64
  netmask ffff:ffff:ffff:ffff::
  gateway 2620:96:c000:1::1
  dns-ip-primary 2620:96:c000:8:e840:68d3:2a77:3d2d
  dns-domain sfblabdm.local
  hip-ip-list 2620:96:c000:1::10
  icmp-address 2620:96:c000:1::10
  ssh-address 10.64.3.163
network-interface
  name wancom1
  pri-utility-addr 169.254.1.1
  sec-utility-addr 169.254.1.2
  netmask 255.255.255.252
network-interface
  name wancom2
  pri-utility-addr 169.254.2.1
  sec-utility-addr 169.254.2.2
  netmask 255.255.255.252
phy-interface
  name s0p0
  operation-type Media
  virtual-mac 00:08:25:04:0d:1e
phy-interface
  name s1p0
  operation-type Media
  slot 1
  virtual-mac 00:08:25:04:0d:1f
phy-interface
  name wancom1
  port 1
  duplex-mode
  speed
  wancom-health-score 8
phy-interface
  name wancom2
  port 2
  duplex-mode
  speed
  wancom-health-score 9
playback-config
  name transferrbt

```

```

entry
    encoding                PCMU
    filename                 US_ringbackPCMU.raw
realm-config
    identifier              SIP-Trunk
    network-interfaces      s0p0:0
    mm-in-realm             enabled
realm-config
    identifier              towards-sfb
    addr-prefix              ::
    network-interfaces      slp0:0
    mm-in-realm             enabled
    refer-call-transfer     enabled
redundancy-config
    peer
        name                Acmesystem1
        type                 Primary
        destination
            address          169.254.1.1:9090
            network-interface wancom1:0
        destination
            address          169.254.2.1:9090
            network-interface wancom2:0
    peer
        name                acmesystem3
        type                 Secondary
        destination
            address          169.254.1.2:9090
            network-interface wancom1:0
        destination
            address          169.254.2.2:9090
            network-interface wancom2:0
response-map
    name                    change183to180
    entries
        rcv-code            183
        xmit-code           180
        reason               Ringing
sdes-profile
    name                    sdes-profile
    egress-offer-format     simultaneous-best-effort
session-agent
    hostname                63.87.147.48
    ip-address              63.87.147.48
    port                    5071
    realm-id                SIP-Trunk
    ping-method             OPTIONS
    ping-interval           30
    out-translationid       stripplus1
session-agent
    hostname                Medpool.sfblabdm.local
    transport-method        StaticTCP
    realm-id                towards-sfb
    ping-method             OPTIONS
    ping-interval           60
    out-translationid       addplus1
    stop-recurse            480
    refer-call-transfer     enabled
    tcp-keepalive           disabled
session-agent
    hostname                med1.sfblabdm.local
    ip-address              172.16.29.44
    port                    5067
    transport-method        StaticTLS

```

realm-id	towards-sfb
ping-method	OPTIONS
ping-interval	60
out-translationid	addplus1
refer-call-transfer	enabled
session-agent	
hostname	med2.sfbldm.local
ip-address	172.16.29.45
port	5067
transport-method	StaticTLS
realm-id	towards-sfb
ping-method	OPTIONS
ping-interval	60
out-translationid	addplus1
refer-call-transfer	enabled
session-timer-profile	
name	towardssfb
session-expires	900
request-refresher	none
session-translation	
id	addplus1
rules-calling	addplus1
rules-called	addplus1
session-translation	
id	stripplus1
rules-calling	stripplus1
rules-called	stripplus1
sip-config	
home-realm-id	towards-sfb
registrar-domain	*
registrar-host	*
options	max-udp-length=0
sip-feature	
name	100rel
realm	SIP-Trunk
require-mode-inbound	Pass
require-mode-outbound	Pass
sip-interface	
realm-id	SIP-Trunk
sip-port	
address	192.65.79.126
out-manipulationid	NATandChangeContact
sip-interface	
realm-id	towards-sfb
sip-port	
address	2620:96:c000:1::10
transport-protocol	TCP
options	100rel-interworking
in-manipulationid	ForEarlyMedia
out-manipulationid	outManipToSFB
session-timer-profile	towardssfb
sip-manipulation	
name	Addpcmat0183
header-rule	
name	check1for83
header-name	@status-line
action	manipulate
comparison-type	pattern-rule
element-rule	
name	isit183
type	status-code
action	sip-manip
comparison-type	pattern-rule
match-value	183


```

new-value                                fix183sdp
sip-manipulation
  name                                    ChangeContact
  header-rule
    name                                  StoreFromnumber
    header-name                            From
    action                                  manipulate
    element-rule
      name                                  StoreFromnumber_er
      type                                  uri-user-only
      action                                  store
  header-rule
    name                                    ChangeContact
    header-name                            Contact
    action                                  manipulate
    element-rule
      name                                  ChangeContact_er
      type                                  uri-user
      action                                  add
      new-value
$StoreFromnumber.$StoreFromnumber_er.$0
sip-manipulation
  name                                    ChangeRecvOnlyToInactive
  header-rule
    name                                  changeSDP
    header-name                            Content-Type
    action                                  manipulate
    msg-type                                reply
    methods                                  INVITE
    element-rule
      name                                  RecvOnlyToInactive
      parameter-name                        application/sdp
      type                                  mime
      action                                  find-replace-all
      comparison-type                       pattern-rule
      match-value                           a=recvonly
      new-value                             a=inactive
sip-manipulation
  name                                    Changeinactosendonly
  description                              Change inactive to sendonly for pstn tran
  header-rule
    name                                  changeSDP
    header-name                            Content-Type
    action                                  manipulate
    msg-type                                request
    methods                                  INVITE
    element-rule
      name                                  inacttosendonly
      parameter-name                        application/sdp
      type                                  mime
      action                                  find-replace-all
      comparison-type                       pattern-rule
      match-value                           a=inactive
      new-value                             a=sendonly
sip-manipulation
  name                                    Check183
  header-rule
    name                                  check183
    header-name                            @status-line
    action                                  manipulate
    comparison-type                       pattern-rule
    element-rule
      name                                  is183
      type                                  status-code

```

```

        action
        comparison-type
        match-value
        new-value
sip-manipulation
    name
    header-rule
        name
        header-name
        action
        msg-type
        methods
    header-rule
        name
        header-name
        action
        msg-type
        methods
        new-value
    header-rule
        name
        header-name
        action
        new-value
    header-rule
        name
        header-name
        action
        new-value
sip-manipulation
    name
    header-rule
        name
        header-name
        action
        comparison-type
        element-rule
            name
            type
            action
            comparison-type
            match-value
            new-value
sip-manipulation
    name
    mime-sdp-rule
        name
        action
        comparison-type
        sdp-media-rule
            name
            media-type
            action
            comparison-type
            sdp-line-rule
                name
                type
                action
                comparison-type
                match-value
        new-value
        action
        comparison-type
        match-value
        new-value
        sip-manip
        pattern-rule
        183
        Modcline
        ForEarlyMedia
        delsupported
        Supported
        delete
        request
        INVITE
        addrequireinINVITE
        Require
        add
        request
        INVITE
        100rel
        Options
        From
        sip-manip
        OptionsResponseLocally
        Change183
        From
        sip-manip
        Check183
        Map183
        check183
        @status-line
        store
        pattern-rule
        is183
        status-code
        replace
        pattern-rule
        183
        180
        Modcline
        test
        manipulate
        pattern-rule
        checkcline
        audio
        manipulate
        pattern-rule
        detectc
        c
        sip-manip
        pattern-rule
        convert183to180
    header-rule
^ (. (?! (2620:0096:c000:1::10|2620:96:c000:8:24b6:215d:7023:ec7d|2620:96:c000:8:55d6:58f5:b47:fe82))) *$

```

```

        name                               checkcl
        header-name                         Content-Length
        action                               sip-manip
        match-value                         0
        new-value                           convert183to180
sip-manipulation
  name                                     NATandChangeContact
  description                             NAT plus Change Contact
  header-rule
    name                                   doNATfortrunk
    header-name                             From
    action                                   sip-manip
    new-value                               NATting
  header-rule
    name                                     delsupported
    header-name                             Supported
    action                                   delete
    msg-type                                request
    methods                                  INVITE
  header-rule
    name                                     addPAIonCallForwards
    header-name                             From
    action                                   sip-manip
    msg-type                                request
    methods                                  INVITE
    new-value                               addPAIonCallForwards
  header-rule
    name                                     addPAIRefers
    header-name                             From
    action                                   sip-manip
    msg-type                                request
    methods                                  INVITE
    new-value                               addPAIRefers
  header-rule
    name                                     ChangeContactHeader
    header-name                             From
    action                                   sip-manip
    new-value                               ChangeContact
  header-rule
    name                                     modify183sdp
    header-name                             From
    action                                   sip-manip
    new-value                               Addpcmato183
sip-manipulation
  name                                     NATting
  header-rule
    name                                     From
    header-name                             From
    action                                   manipulate
    msg-type                                request
    element-rule
      name                                   From_header
      type                                   uri-host
      action                                   replace
      new-value                               $LOCAL_IP
  header-rule
    name                                     To
    header-name                             To
    action                                   manipulate
    msg-type                                request
    element-rule
      name                                   To
      type                                   uri-host
      action                                   replace

```

```

new-value                                $REMOTE_IP

sip-manipulation
  name                                     OptionsResponseLocally
  header-rule
    name                                   rejectOptions
    header-name                             request-uri
    action                                   reject
    msg-type                                 request
    methods                                  OPTIONS
    new-value                               200: OK

sip-manipulation
  name                                     addPAIRefers
  header-rule
    name                                     checkReferredBy
    header-name                             REFERRED-BY
    action                                   store
    msg-type                                 request
    methods                                  INVITE
  header-rule
    name                                     addPAIifReferredBy
    header-name                             P-Asserted-Identity
    action                                   add
    comparison-type                         pattern-rule
    msg-type                                 request
    methods                                  INVITE
    match-value                             $checkReferredBy
    new-value                               <sip:5712935325+@+$LOCAL_IP>

sip-manipulation
  name                                     addPAIonCallForwards
  header-rule
    name                                     checkForHistoryInfo
    header-name                             History-Info
    action                                   store
    msg-type                                 request
    methods                                  INVITE
  header-rule
    name                                     addPAIifHistoryInfo
    header-name                             P-Asserted-Identity
    action                                   add
    comparison-type                         pattern-rule
    msg-type                                 request
    methods                                  INVITE
    match-value                             $checkForHistoryInfo
    new-value                               <sip:5712935325+@+$LOCAL_IP>

sip-manipulation
  name                                     convert183to180
  header-rule
    name                                     delSDP
    header-name                             Content-Type
    action                                   manipulate
    comparison-type                         case-insensitive
    element-rule
      name                                   del183SDP
      parameter-name                         application/sdp
      type                                    mime
      action                                  delete-element
  header-rule
    name                                     delContentType
    header-name                             Content-Type
    action                                   manipulate
    element-rule
      name                                   delCT
      parameter-name                         *
      type                                    header-param

```

```

        action
        delete-header
header-rule
    name
        change487to486
    header-name
        @status-line
    action
        manipulate
    element-rule
        name
            modStatusCode
        type
            status-code
        action
            replace
        match-value
            183
        new-value
            180
    element-rule
        name
            modReasonPhrase
        type
            reason-phrase
        action
            replace
        comparison-type
            case-insensitive
        match-value
            Session Progress
        new-value
            Ringing
sip-manipulation
    name
        fix183sdp
    description
        To bypass the reinvoke
    header-rule
        name
            Checkforpcma
        header-name
            Content-type
        action
            store
        element-rule
            name
                Checkpcmaexists
            parameter-name
                application/sdp
            type
                mime
            action
                store
            match-value
                (a=rtpmap:8 PCMA/8000)
    header-rule
        name
            Addpcma
        header-name
            Content-Type
        action
            manipulate
        comparison-type
            boolean
        match-value
            !$Checkforpcma.$Checkpcmaexists
        element-rule
            name
                Addpcma
            parameter-name
                application/sdp
            type
                mime
            action
                find-replace-all
            comparison-type
                pattern-rule
            match-value
                a=rtpmap:0 PCMU/8000
            new-value
                "a=rtpmap:0
PCMU/8000"+$CRLF+"a=rtpmap:8 PCMA/8000"
    mime-sdp-rule
        name
            modmline
        action
            manipulate
        sdp-media-rule
            name
                modmline_m
            media-type
                audio
            action
                manipulate
            sdp-line-rule
                name
                    change_payload
                type
                    m
                action
                    find-replace-all
                comparison-type
                    pattern-rule
                match-value
                    ^(audio) ( [0-
9]{4,5}) ( RTP/AVP 0 101)$
                new-value
                    audio+$2+" RTP/AVP
0 8 101"
    sip-manipulation
        name
            modcontact

```

```

header-rule
  name                               modcontact_er
  header-name                         Contact
  action                              manipulate
  comparison-type                     pattern-rule
  element-rule
    name                               moder
    type                               uri-user
    action                             replace
    comparison-type                   pattern-rule
    match-value                       ((\+1)(\+1\d+))
    new-value                          $3
  element-rule
    name                               modanon
    type                               uri-user
    action                             replace
    comparison-type                   pattern-rule
    match-value                       \+1anonymous
    new-value                          anonymous
sip-manipulation
  name                               outManipToSFB
  header-rule
    name                               From
    header-name                       From
    action                             manipulate
    msg-type                           request
    element-rule
      name                               From_header
      type                               uri-host
      action                             replace
      new-value                          oracle.sfblabdm.local
    element-rule
      name                               modanonfrom
      type                               uri-user
      action                             replace
      comparison-type                   pattern-rule
      match-value                       \+1anonymous
      new-value                          anonymous
  header-rule
    name                               To
    header-name                       To
    action                             manipulate
    msg-type                           request
    element-rule
      name                               remplusone
      type                               uri-user
      action                             replace
      comparison-type                   pattern-rule
      match-value                       \+1$
    element-rule
      name                               To
      type                               uri-host
      action                             replace
      new-value                          medpool.sfblabdm.local
  header-rule
    name                               fixcontact
    header-name                       Contact
    action                             manipulate
    element-rule
      name                               updatecontact
      type                               uri-host
      action                             replace
      new-value                          oracle.sfblabdm.local
header-rule

```

name	fixcontactuser
header-name	Contact
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
element-rule	
name	user
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	(.*)
new-value	"+"+\$ORIGINAL
element-rule	
name	moder
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	((\+1) (\+1\d+))
new-value	\$3
element-rule	
name	modanoncon
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	\+1anonymous
new-value	anonymous
header-rule	
name	Ruri
header-name	Request-URI
action	manipulate
msg-type	request
element-rule	
name	remplusone
type	uri-user
action	replace
comparison-type	pattern-rule
match-value	\+1\$
header-rule	
name	addReplaces
header-name	Supported
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
new-value	\$ORIGINAL+, replaces
header-rule	
name	addUpdate
header-name	Allow
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
new-value	\$ORIGINAL+, UPDATE
header-rule	
name	addingptime
header-name	From
action	sip-manip
msg-type	request
new-value	replacemptimewithptime
sip-manipulation	
name	replacemptimewithptime
mime-sdp-rule	
name	addptime

```

msg-type request
action manipulate
sdp-media-rule
  name addSDPptime
  media-type audio
  action manipulate
  sdp-line-rule
    name addptimelr
    type a
    action find-replace-all
    match-value maxptime:20
    new-value ptime:20
sip-manipulation
  name test183
  header-rule
    name manipulate183
    header-name content-type
    action manipulate
    comparison-type pattern-rule
    msg-type reply
    methods INVITE
    element-rule
      name check183_er
      parameter-name application/sdp
      type mime
      action sip-manip
      comparison-type pattern-rule
      match-value (.*) (?!c=IN IP4
172.16.29.44) (.*)
      new-value convert183to180
sip-monitoring
  match-any-filter enabled
  monitoring-filters all,ip6
spl-config
steering-pool
  ip-address 192.65.79.126
  start-port 49152
  end-port 65535
  realm-id SIP-Trunk
steering-pool
  ip-address 2620:96:c000:1::10
  start-port 49152
  end-port 65535
  realm-id towards-sfb
system-config
  hostname ACMESYSTEM
  description 3820 Skype for Business IOT
  default-gateway 192.65.79.33
  source-routing enabled
  default-v6-gateway 2620:96:c000:1::1
tls-global
  session-caching enabled
tls-profile
  name sfb-tls-profile
  end-entity-certificate SBCcert1
  trusted-ca-certificates CACert
  mutual-authenticate enabled
  tls-version tlsv12
translation-rules
  id addplus1
  type add
  add-string +1
translation-rules
  id stripplus1

```




type	delete
delete-string	+1

Test Plan Executed

Following is the test plan executed against this setup and results have been documented below.

Test case ID	TCP/TLS	Title	Result	Comments
408058	TCP	PstnEndPt calls LyncEndPt with Caller ID set to 'Anonymous' on DUT	Pass	
408282	TLS	DUT accepts MedSrv 'pool' certificate for a secure call	Pass	
428361	TLS	DUT offers DUT pool certificate for a secure call	Pass	
408064	TCP	LyncEndPt calls IVR number and navigates through the IVR menu before call Connection	Pass	
408106	TCP	LyncEndPt hears Early Media for a call to PstnEndPt	Pass	
408112	TCP	DUT sends PRACK for reliable Early Media for a call from PstnEndPt to LyncEndPt	Pass	
408079	TCP	PstnEndPt1 calls LyncEndPt that is set to simultaneous ring to LyncEndPt and PstnEndPt2 answers	Pass	
408109	TCP	DUT disconnects a forked call if PstnEndPt hangs up while phones are ringing	Pass	
408159	TCP	DUT disconnects a forked call if PstnEndPt hangs up while phones are ringing. (Media Bypass OFF) (IPv6)	Pass	
408062	TCP	LyncEndPt calls PstnEndPt with a call duration longer than 32 seconds	Pass	
408063	TCP	DUT generates 603 response for a call rejected by PstnEndPt	Pass	
408065	TCP	DUT correctly handles non-E.164 number in outbound Request URI	Pass	
408066	TCP	LyncEndPt calls PstnEndPt and hangs up before receiving 200 from DUT	Pass	
408067	TCP	PstnEndPt displays LyncEndPt Caller ID for Outbound Call	Pass	
408068	TCP	PstnEndPt disconnects established call to LyncEndPt	Pass	
408069	TCP	PstnEndPt disconnects established call from LyncEndPt	Pass	

408071	TCP	DUT processes phone-context in Request and To URI from LyncEndPt	Pass	
408072	TCP	LyncEndPt sends INVITE with E.164 number and extension in Request and To URI	Pass	
408073	TCP	DUT processes call from LyncEndPt with E.164 number in FROM Header URI	Pass	
408074	TCP	LyncEndPt response to PstnEndPt is delayed due to network delay	Pass	
408075	TCP	DUT is able to disconnect a call that is forked to LyncEndPts set to 'Do not disturb'	Pass	
408076	TCP	DUT sends single media description line for a call from PstnEndPt to LyncEndPt	Pass	
408077	TCP	LyncEndPt calls an IVR number and navigates through the IVR menu after call connection.	Pass	
408078	TCP	DUT handles call from MedSrv with an alias name in the FROM header	Pass	
408081	TCP	MedSrv renegotiates an existing voice session with a different IP address	Pass	
408082	TCP	PstnEndPt calls LyncEndPt1, LyncEndPt1 parks the call and retrieves it on LyncEndPt2	Pass	
408085	TCP	DUT establishes call to LyncEndPt with configured value of ptime	Pass	
408098	TLS	LyncEndPt makes a secure call to an IVR number and navigates through the IVR menu after receiving 200 from DUT	Pass	
408116	TLS	DUT that supports SRTP only rejects call from LyncEndPt that supports RTP Only	Pass	
433140	TLS	LyncEndPt makes a secure call to an IVR and pastes a string of conference ID digits which are recognized by the DUT and IVR	Pass	
408086	TCP	LyncEndPt makes a call to PstnEndPt with G.711 A-law and/or G.711 U-law codecs	Pass	
408090	TCP	PstnEndPt is able to establish a call with LyncEndPt using G.711 A-law	Pass	

		codec		
408092	TCP	LyncEndPt is able to establish a call with PstnEndPt using G.711 A-law codec	Pass	
408101	TCP	DUT offers DTMF payload type in the range of 96-127 to MedSrv	Pass	
408114	TCP	LyncEndPt makes a call to PstnEndPt with G.711 U-law codec	Pass	
408119	TCP	LyncEndPt receives a call from PstnEndPt with G.711 U-law codecs	Pass	
408093	TCP	DUT handles multiple RTP streams for a call to LyncEndPt	Pass	
408100	TCP	DUT does not change the SSRC of an established outbound RTP session	Pass	
408104	TCP	DUT does not change the SSRC of an established inbound RTP session	Pass	
408123	TLS	DUT does not change the SSRC of an established outbound SRTP session	Pass	
408126	TLS	DUT does not change the SSRC of an established inbound SRTP session	Pass	
408091	TLS	PstnEndPt is able to establish a secure call with LyncEndPt using G.711 A-law codec	Pass	
408094	TLS	DUT handles multiple SRTP streams for a secure call to LyncEndPt	Pass	
408107	TLS	LyncEndPt hears Early Media for a secure call to PstnEndPt	Pass	
408108	TLS	LyncEndPt hears Early Media for a secure call to PstnEndPt when Media Bypass OFF	Pass	
408110	TLS	DUT disconnects a forked secure call if PstnEndPt hangs up while phones are ringing	Pass	
408120	TLS	LyncEndPt receives a secure call with G.711 U-law codec with Media Bypass OFF	Pass	
408122	TLS	LyncEndPt makes a secure call to PstnEndPt and PstnEndPt later hangs up	Pass	
408129	TLS	LyncEndPt makes a secure call to PstnEndPt	Pass	
408160	TLS	DUT disconnects a forked secure call if PstnEndPt hangs up while phones	Pass	

		are ringing. (Media Bypass OFF) (IPv6)		
428537	TLS	LyncEndPt places a secure call to PstnEndPt and call is up for more than 30 minutes with session timer enabled on DUT	Pass	
439173	TLS	LyncEndPt places a secure call to PstnEndPt and call is up for more than 30 minutes with session timer enabled on DUT (Media Bypass OFF)	Pass	
408080	TCP	Inbound call to LyncEndPt from PstnEndPt with a very long Request-URI in the INVITE	Pass	
408083	TCP	LyncEndPt1 calls PstnEndPt, LyncEndPt1 parks the call and retrieves it on LyncEndPt2	Pass	
408084	TCP	PstnEndPt calls LyncEndPt that later parks the call but does not retrieve it	Pass	
408111	TCP	PstnEndPt1 calls LyncEndPt that is set to simultaneous ring to PstnIVR	Pass	
408115	TLS	MedSrv that requires SRTP rejects call from DUT that supports RTP only	Pass	
408117	TLS	DUT handles 488 response from the MedSrv operating in RTP only mode	Pass	
408118	TLS	DUT sends its own FQDN in contact header for TLS call from LyncEndPt to PstnEndPt	Pass	
408125	TLS	PstnEndPt calls LyncEndPt with security enabled and LyncEndPt later hangs up	Pass	
408127	TLS	DUT adds at least one "crypto" attribute for each media description line in the SDP	Pass	
408148	TCP	PstnEndPt1 calls LyncEndPt that is set to simultaneous ring to LyncEndPt and PstnEndPt2 answers. (Media Bypass OFF) (IPv6)	Pass	
408153	TCP	PstnEndPt calls LyncEndPt 1, LyncEndPt 1 parks the call and retrieves it on LyncEndPt 2. (Media Bypass OFF) (IPv6)	Pass	
408155	TCP	PstnEndPt calls LyncEndPt that later parks the call but does not retrieve it. (Media Bypass OFF) (IPv6)	Pass	
408162	TCP	DUT sends PRACK for reliable Early	Pass	

		Media for a call from PstnEndPt to LyncEndPt. (Media Bypass OFF) (IPv6)		
408205	TLS	PstnEndPt1 makes a secure call to LyncEndPt that has call forwarded to PstnEndPt2	Pass	
408206	TLS	PstnEndPt1 makes a secure call to LyncEndPt that forwards the call to PstnEndPt2 with Media Bypass OFF	Pass	
408207	TCP	PstnEndPt1 calls LyncEndPt that forwards all calls to PstnEndPt2 when Media Bypass OFF	Pass	
408213	TCP	LyncEndPt1 calls LyncEndPt2 and escalates the call to a conference, inviting PstnEndPt and later removing it	Pass	
408214	TCP	PstnEndPt establishes a call with the Conference Auto Attendant	Pass	
408216	TCP	LyncEndPt1 calls LyncEndPt2 and escalates the call to a conference, inviting PstnEndPt and later removing it. (Media Bypass OFF) (IPv6)	Pass	
408227	TCP	LyncEndPt resumes call to PstnEndPt after playing music on hold for 15 minutes	Pass	
408229	TCP	LyncEndPt places a call to PstnEndPt on hold and resumes after 12 minutes	Pass	
428506	TLS	LyncEndPt places secure call to PstnEndPt on hold after 30 minutes and then resumes	Pass	
408224	TLS	PstnEndPt places a secure call to LyncEndPt on hold and resumes after 15 minutes	Pass	
408235	TLS	PstnEndPt places a secure call from LyncEndPt on hold and then resumes	Pass	
408231	TCP	LyncEndPt plays music when it holds call from PstnEndPt to LyncEndPt	Pass	
408261	TCP	PstnEndPt1 calls LyncEndPt and LyncEndPt Consultative Transfers to PstnEndPt2	Pass	
408262	TLS	PstnEndPt1 makes a secure call to LyncEndPt and LyncEndPt Consultative Transfers to PstnEndPt2	Pass	
408263	TCP	DUT does not drop the call when	Pass	

		Consultative Transfer by LyncEndPt to second PstnEndPt fails		
408264	TCP	DUT supports Hairpin Elimination for Consultative Transfer with REFER	Pass	
408265	TLS	DUT supports Hairpin Elimination for secure Consultative Transfer with REFER	Pass	
408275	TLS	PstnEndPt1 makes a secure call to LyncEndPt and LyncEndPt Consultative Transfers to PstnEndPt2. (Media Bypass OFF) (IPv6)	Pass	
408254	TCP	DUT includes REFER in ALLOW header in INVITE sent to MedSrv	Pass	
408255	TCP	PstnEndPt1 calls LyncEndPt and LyncEndPt Blinds Transfers the call to PstnEndPt2	Pass	
408256	TLS	PstnEndPt1 makes a secure call to LyncEndPt and LyncEndPt Blinds Transfers the call to PstnEndPt2	Pass	
408259	TLS	DUT maintains the original session when rejecting a call transfer with REFER	Pass	
408268	TCP	PstnEndPt1 makes a secure call to LyncEndPt and LyncEndPt Blinds Transfers the call to PstnEndPt2. (Media Bypass OFF) (IPv6)	Pass	
408273	TCP	PstnEndPt calls LyncEndPt1 that performs Blind Transfer to LyncEndPt2 with REFER. (Media Bypass OFF) (IPv6)	Pass	
408285	TLS	DUT uses load balancing to distribute secure inbound calls among MedSrvs in a cluster	Pass	
408286	TCP	DUT responds to OPTIONS as keep alive to MedSrv over TCP	Pass	
408287	TCP	DUT sends periodic OPTIONS message as keep alive to MedSrv	Pass	
408288	TLS	DUT responds to OPTIONS as keep alive to MedSrv over TLS	Pass	
408289	TCP	DUT resumes sending calls to MedSrv when it starts receiving OPTIONS response from that MedSrv	Pass	
408291	TCP	PstnEndPt establishes a call with LyncEndPt when interface of MedSrv1 goes down	Pass	

408293	TCP	DUT fails over incoming call to a second MedSrv when the first MedSrv does not respond	Pass	
408306	TCP	DUT utilizes failover and does not offer new calls to a failed MedSrv	Pass	
408309	TCP	DUT distributes new calls among DNS configured MedSrvs	Pass	
408311	TCP	DUT honors TTL when distributing new calls among DNS configured MedSrvs	Pass	
408315	TCP	DUT sends 414 when unable to handle very long Request URI	Pass	
408317	TCP	DUT processes 488 response for unsupported codec from MedSrv	Pass	
408321	TCP	DUT disconnects call when MedSrv sends 408 for call from PstnEndPt	Pass	
408322	TLS	DUT processes 603 from LyncEndPt for a secure call	Pass	
408323	TCP	DUT processes 603 response from LyncEndPt	Pass	
408324	TCP	DUT handles call from LyncEndPt to a user that does not exist in the domain	Pass	
408325	TCP	DUT generates 486 response from a busy PstnEndPt	Pass	
408326	TCP	DUT processes 486 response from a busy LyncEndPt	Pass	
408327	TCP	DUT disconnects call when MedSrv sends 501 for call from PstnEndPt	Pass	
408328	TCP	DUT disconnects call when MedSrv sends 606 for call from PstnEndPt	Pass	
408329	TCP	DUT responds with 488 when MedSrv offers a codec unsupported on the device	Pass	
408347	TCP	LyncEndPt receives a call from PstnEndPt with G.711 A-law and/or G.711 U-law codecs	Pass	
408348	TCP	PstnEndPt1 calls LyncEndPt that forwards the call to PstnEndPt2	Pass	
408349	TCP	PstnEndPt1 calls LyncEndPt that escalates the call to a conference by inviting PstnEndPt2	Pass	
408350	TCP	DUT fails over incoming call to	Pass	

		MedSrv2 when MedSrv1 sends 503 response		
--	--	---	--	--

Troubleshooting Tools

If you find that you are not able to complete calls or have problems with the test cases, there are a few tools available for Windows Server, Lync Server, and the Oracle SBC like logging and tracing which may be of assistance. In this section we will provide a list of tools which you can use to aid in troubleshooting any issues you may encounter.

Since we are concerned with communication between the Lync Server mediation server and the SBC we will focus on the troubleshooting tools to use between those devices if calls are not working or tests are not passing.

Microsoft Network Monitor (NetMon)

NetMon is a network protocol analyzer which is freely downloadable from Microsoft. It can be found at www.microsoft.com/downloads. NetMon could be installed on the Lync Server mediation server, the Lync Server Standard Edition server, or Enterprise Edition front end server.

Wireshark

Wireshark is also a network protocol analyzer which is freely downloadable from www.wireshark.org. Wireshark could be installed on the Lync Server mediation server, the Lync Server Standard Edition server, or MCS Enterprise Edition front end server.

Eventviewer

There are several locations in the event viewer where you can find valuable information to aid in troubleshooting issues with your deployment.

With the requirement that there is a completely functioning Lync Server with Enterprise Voice deployment in place, there are only a few areas in which one would use the Event Viewer for troubleshooting:

- The Enterprise Voice client;
- The Lync Server Front End server;
- A Lync Server Standard Edition Server; and
- A Lync Server Mediation Server.

On the Oracle SBC 4600 Series

The Oracle SBC provides a rich set of statistical counters available from the ACLI, as well as log file output with configurable detail. The follow sections detail enabling, adjusting and accessing those interfaces.

Resetting the statistical counters, enabling logging and restarting the log files.

At the SBC Console:

```
oraclesbc1# reset sipd
oraclesbc1# notify sipd debug
oraclesbc1#
enabled SIP Debugging
oraclesbc1# notify all rotate-logs
```

Examining the log files

Note: You will FTP to the management interface of the SBC with the username user and user mode password (the default is "acme").

```
C:\Documents and Settings\user>ftp 192.168.5.24
Connected to 192.168.85.55.
220 oraclesbc1FTP server (VxWorks 6.4) ready.
User (192.168.85.55:(none)): user
331 Password required for user.
Password: acme
230 User user logged in.
ftp> cd /ramdrv/logs
250 CWD command successful.
ftp> get sipmsg.log
200 PORT command successful.
150 Opening ASCII mode data connection for '/ramdrv/logs/sipmsg.log' (3353
bytes).
226 Transfer complete.
ftp: 3447 bytes received in 0.00Seconds 3447000.00Kbytes/sec.
ftp> get log.sipd
200 PORT command successful.
150 Opening ASCII mode data connection for '/ramdrv/logs/log.sipd' (204681
bytes).
226 Transfer complete.
ftp: 206823 bytes received in 0.11Seconds 1897.46Kbytes/sec.
ftp> bye
221 Goodbye.
```

You may now examine the log files with the text editor of your choice.

Through the Web GUI

You can also check the display results of filtered SIP session data from the Oracle Enterprise Session Border Controller, and provides traces in a common log format for local viewing or for exporting to your PC. Please check the “Monitor and Trace” section (page 145) of the Web GUI User Guide available at http://docs.oracle.com/cd/E56581_01/index.htm

Telnet

Since we are working within an architecture which uses bound TCP listening ports for functionality, the simplest form of troubleshooting can be seeing if the devices are listening on a particular port, as well as confirming that there is nothing blocking them such as firewalls. Ensure that you have a TELNET client available on a workstation as well as on the Lync Server mediation server.

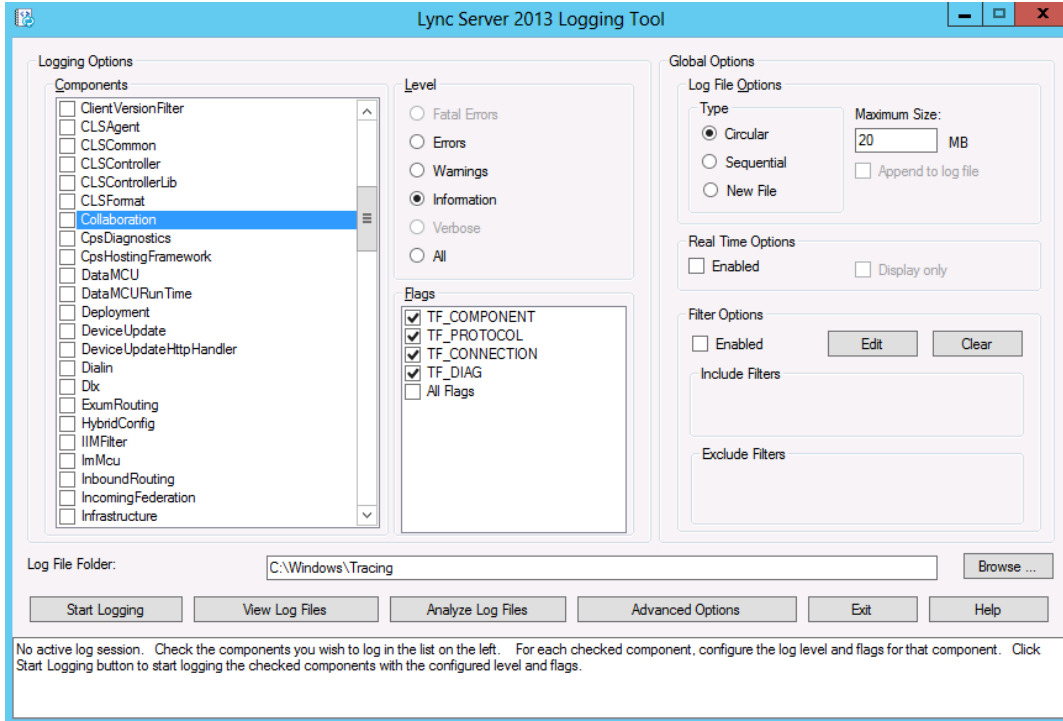
The Lync Server mediation server will listen on TCP port 5067 by default for SIP signaling. In our example we are listening on 5060 on the PSTN facing NIC. From the Standard Edition pool or Enterprise Edition pool the Mediation Server would be listening on port 5061. Tests may include:

- Client to pool server: **telnet <servername> 5061**
- Pool server to Mediation Server: **telnet <servername> 5061**

On the Lync Server

Lync Server Logging Tool

The Skype for Business Logging Tool provides internal traces and messaging between different Skype for Business elements like Front-end, Mediation server, Lync Clients, etc. File name is OCSReskit.msi. Once installed, it can be accessed from any one of the Lync Server servers by running Start/Microsoft Skype for Business/Lync Server Logging Tool.



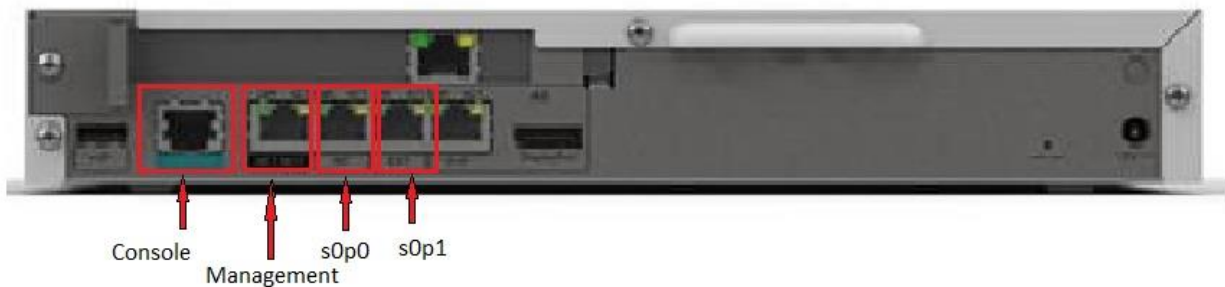
Appendix A

Accessing the ACLI

Access to the ACLI is provided by:

- The serial console connection;
- TELNET, which is enabled by default but may be disabled; and
- SSH, this must be explicitly configured.

Initial connectivity will be through the serial console port. At a minimum, this is how to configure the management (eth0) interface on the SBC.

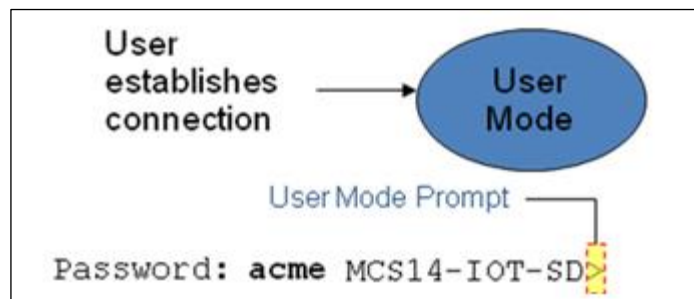


ACLI Basics

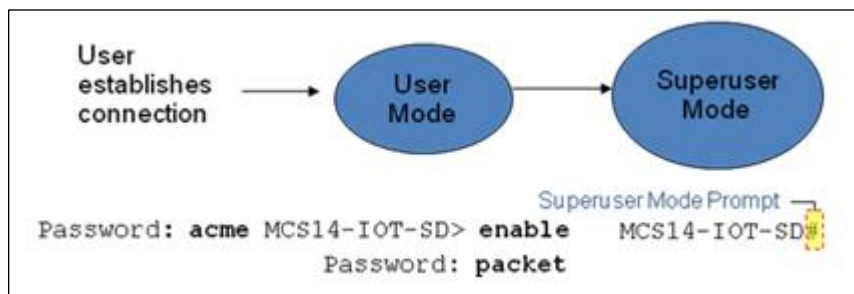
There are two password protected modes of operation within the ACLI, User mode and Superuser mode.

When you establish a connection to the SBC, the prompt for the User mode password appears. The default password is acme.

User mode consists of a restricted set of basic monitoring commands and is identified by the greater than sign (>) in the system prompt after the target name. You cannot perform configuration and maintenance from this mode.



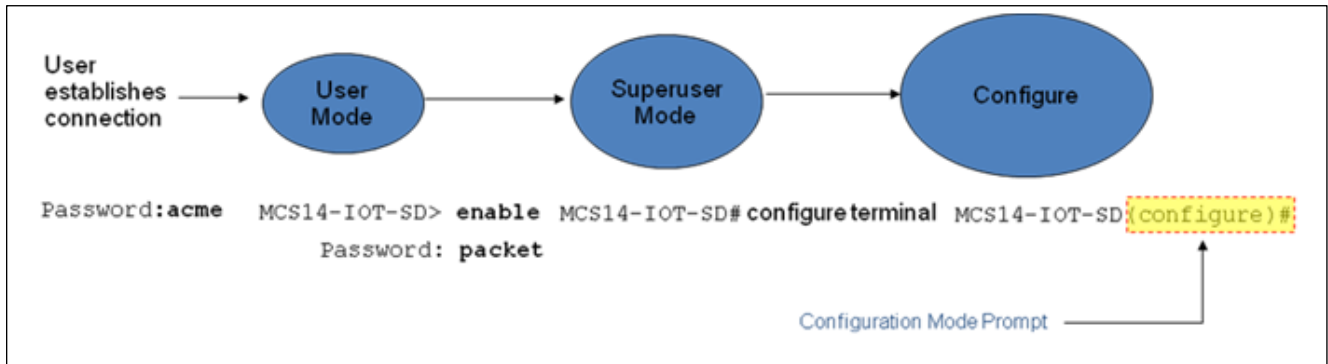
The Superuser mode allows for access to all system commands for operation, maintenance, and administration. This mode is identified by the pound sign (#) in the prompt after the target name. To enter the Superuser mode, issue the enable command in the User mode.



From the Superuser mode, you can perform monitoring and administrative tasks; however you cannot configure any elements. To return to User mode, issue the exit command.

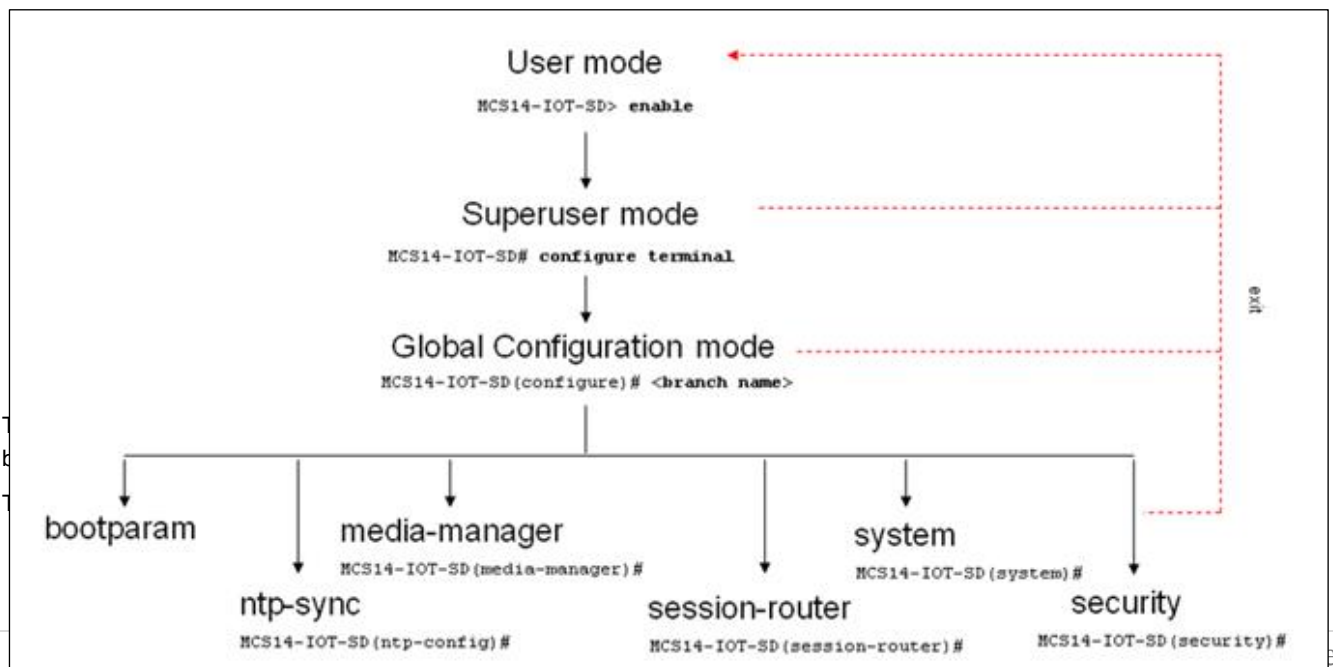
You must enter the Configuration mode to configure elements. For example, you can access the configuration branches and configuration elements for signaling and media configurations. To enter the Configuration mode, issue the **configure terminal** command in the Superuser mode.

Configuration mode is identified by the word configure in parenthesis followed by the pound sign (#) in the prompt after the target name, for example, **oraclesbc1(configure)#**. To return to the Superuser mode, issue the **exit** command.



In the configuration mode, there are six configuration branches:

- bootparam;
- ntp-sync;
- media-manager;
- session-router;
- system; and
- security.



- boot device – The global management port, usually eth0
- file name – The boot path and the image file.
- inet on ethernet – The IP address and subnet mask (in hex) of the management port of the SD.
- host inet –The IP address of external server where image file resides.
- user and ftp password – Used to boot from the external FTP server.
- gateway inet – The gateway IP address for reaching the external server, if the server is located in a different network.

```

'.' = clear field; '-' = go to previous field; q = quit
boot device          : eth0
processor number     : 0
host name            :
file name            : /tffs0/nnSCX620.gz
inet on ethernet (e) : 10.0.3.11:ffff0000
inet on backplane (b) :
host inet (h)        : 10.0.3.100
gateway inet (g)     : 10.0.0.1
user (u)             : anonymous
ftp password (pw) (blank = rsh) : anonymous
flags (f)            : 0x8
target name (tn)     : MCS14-IOT-SD
startup script (s)   :
other (o)

```

The ntp-sync branch provides access to ntp server configuration commands for synchronizing the SBC time and date.

The security branch provides access to security configuration.

The system branch provides access to basic configuration elements as system-config, snmp-community, redundancy, physical interfaces, network interfaces, etc.

The session-router branch provides access to signaling and routing related elements, including H323-config, sip-config, ivf-config, local-policy, sip-manipulation, session-agent, etc.

The media-manager branch provides access to media-related elements, including realms, steering pools, dns-config, media-manager, and so forth.

You will use media-manager, session-router, and system branches for most of your working configuration.

Configuration Elements

The configuration branches contain the configuration elements. Each configurable object is referred to as an element. Each element consists of a number of configurable parameters.

Some elements are single-instance elements, meaning that there is only one of that type of the element - for example, the global system configuration and redundancy configuration.

Some elements are multiple-instance elements. There may be one or more of the elements of any given type. For example, physical and network interfaces.

Some elements (both single and multiple instance) have sub-elements. For example:

- SIP-ports - are children of the sip-interface element
- peers – are children of the redundancy element
- destinations – are children of the peer element

Creating an Element

1. To create a single-instance element, you go to the appropriate level in the ACLI path and enter its parameters. There is no need to specify a unique identifier property because a single-instance element is a global element and there is only one instance of this element.
2. When creating a multiple-instance element, you must specify a unique identifier for each instance of the element.
3. It is important to check the parameters of the element you are configuring before committing the changes. You do this by issuing the `show` command before issuing the `done` command. The parameters that you did not configure are filled with either default values or left empty.
4. On completion, you must issue the `done` command. The done command causes the configuration to be echoed to the screen and commits the changes to the volatile memory. It is a good idea to review this output to ensure that your configurations are correct.
5. Issue the `exit` command to exit the selected element.

Note that the configurations at this point are not permanently saved yet. If the SBC reboots, your configurations will be lost.

Editing an Element

The procedure of editing an element is similar to creating an element, except that you must select the element that you will edit before editing it.

1. Enter the element that you will edit at the correct level of the ACLI path.
2. Select the element that you will edit, and view it before editing it.
The `select` command loads the element to the volatile memory for editing. The `show` command allows you to view the element to ensure that it is the right one that you want to edit.
3. Once you are sure that the element you selected is the right one for editing, edit the parameter one by one. The new value you provide will overwrite the old value.
4. It is important to check the properties of the element you are configuring before committing it to the volatile memory. You do this by issuing the `show` command before issuing the `done` command.
5. On completion, you must issue the `done` command.
6. Issue the `exit` command to exit the selected element.

Note that the configurations at this point are not permanently saved yet. If the SBC reboots, your configurations will be lost.

Deleting an Element

The `no` command deletes an element from the configuration in editing.

To delete a single-instance element,

1. Enter the `no` command from within the path for that specific element
2. Issue the `exit` command.

To delete a multiple-instance element,

1. Enter the `no` command from within the path for that particular element.
The key field prompt, such as `<name>:<sub-port-id>`, appears.
2. Use the `<Enter>` key to display a list of the existing configured elements.
3. Enter the number corresponding to the element you wish to delete.
4. Issue the `select` command to view the list of elements to confirm that the element was removed.

Note that the configuration changes at this point are not permanently saved yet. If the SBC reboots, your configurations will be lost.

Configuration Versions

At any time, three versions of the configuration can exist on the SBC: the edited configuration, the saved configuration, and the running configuration.

- The **edited configuration** – this is the version that you are making changes to. This version of the configuration is stored in the SBC's volatile memory and will be lost on a reboot.
To view the editing configuration, issue the `show configuration` command.
- The **saved configuration** – on issuing the `save-config` command, the edited configuration is copied into the non-volatile memory on the SBC and becomes the saved configuration. Because the saved configuration has not been activated yet, the changes in the configuration will not take effect. On reboot, the last activated configuration (i.e., the last running configuration) will be loaded, not the saved configuration.
- The **running configuration** is the saved then activated configuration. On issuing the `activate-config` command, the saved configuration is copied from the non-volatile memory to the volatile memory. The saved configuration is activated and becomes the running configuration. Although most of the configurations can take effect once being activated without reboot, some configurations require a reboot for the changes to take effect.
To view the running configuration, issue command `show running-config`.

Saving the Configuration

The `save-config` command stores the edited configuration persistently.

Because the saved configuration has not been activated yet, changes in configuration will not take effect. On reboot, the last activated configuration (i.e., the last running configuration) will be loaded. At this stage, the saved configuration is different from the running configuration.

Because the saved configuration is stored in non-volatile memory, it can be accessed and activated at later time.

Upon issuing the `save-config` command, the SBC displays a reminder on screen stating that you must use the `activate-config` command if you want the configurations to be updated.

```
oraclesbcl # save-config
Save-Config received, processing.
waiting 1200 for request to finish
Request to 'SAVE-CONFIG' has Finished,
Save complete
Currently active and saved configurations do not match!
To sync & activate, run 'activate-config' or 'reboot activate'.
oraclesbcl #
```


Activating the Configuration





On issuing the `activate-config` command, the saved configuration is copied from the non-volatile memory to the volatile memory. The saved configuration is activated and becomes the running configuration.

Some configuration changes are service affecting when activated. For these configurations, the SBC warns that the change could have an impact on service with the configuration elements that will potentially be service affecting. You may decide whether or not to continue with applying these changes immediately or to apply them at a later time.

```
oraclesbcl# activate-config
Activate-Config received, processing.
waiting 120000 for request to finish
Request to 'ACTIVATE-CONFIG' has Finished,
Activate Complete
oraclesbcl#
```



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