

How to Register JIO SIP Trunk with NENX (SARVAM UCS SOHO)?

Telecom

What
When
Which
Where
How
Who
Why



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Version: V1R2

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SARVAM UCS SOHO is Unified Communication Server on Matrix ETERNITY NENX. Eternity NENX is designed for small and home offices and nascent enterprises. Multiple SIP PRI can be terminated to SARVAM UCS SOHO.

- This document describes the steps to register an JIO SIP Trunk with NENX.

Prerequisites:

- ETERNITY NENX with NX DBM VOCODER64 VoIP Module.
- Eternity NENX with SARVAM UCS SOHO license.

View Profile

Service Profile	As per System
SARVAM UCS SOHO	Yes
SIP Extensions	50
VARTA Essential Users	40
VARTA Professional Users	5
VARTA Collaboration Users	5
Voice Mail	Yes
SMS Server	Yes
CTI	Yes
SMS Gateway	Yes

Close

SARVAM UCS SOHO (ETERNITY NENX) with Eternity NE VS (VoIP Server Module). It is an optional module, requiring license activation.

NOTE : By default, the VoIP module is factory fitted in ETERNITY NENXIP50. However, it is an optional module in ETERNITY NENX312 and ETERNITY NENX416. You must purchase the VoIP module to register and configure SIP trunk and SIP extension.

 SARVAM UCS

Basic Settings ▶

Advanced Settings ▶

Maintenance ▶

Status ▼

→ System Details

→ System Usage

→ System Performance

→ Mailbox Status

→ USB Status

→ CO

→ Mobile

→ SIP Trunk

→ SIP Extension

→ SMS Gateway

→ Network

→ DSS Status

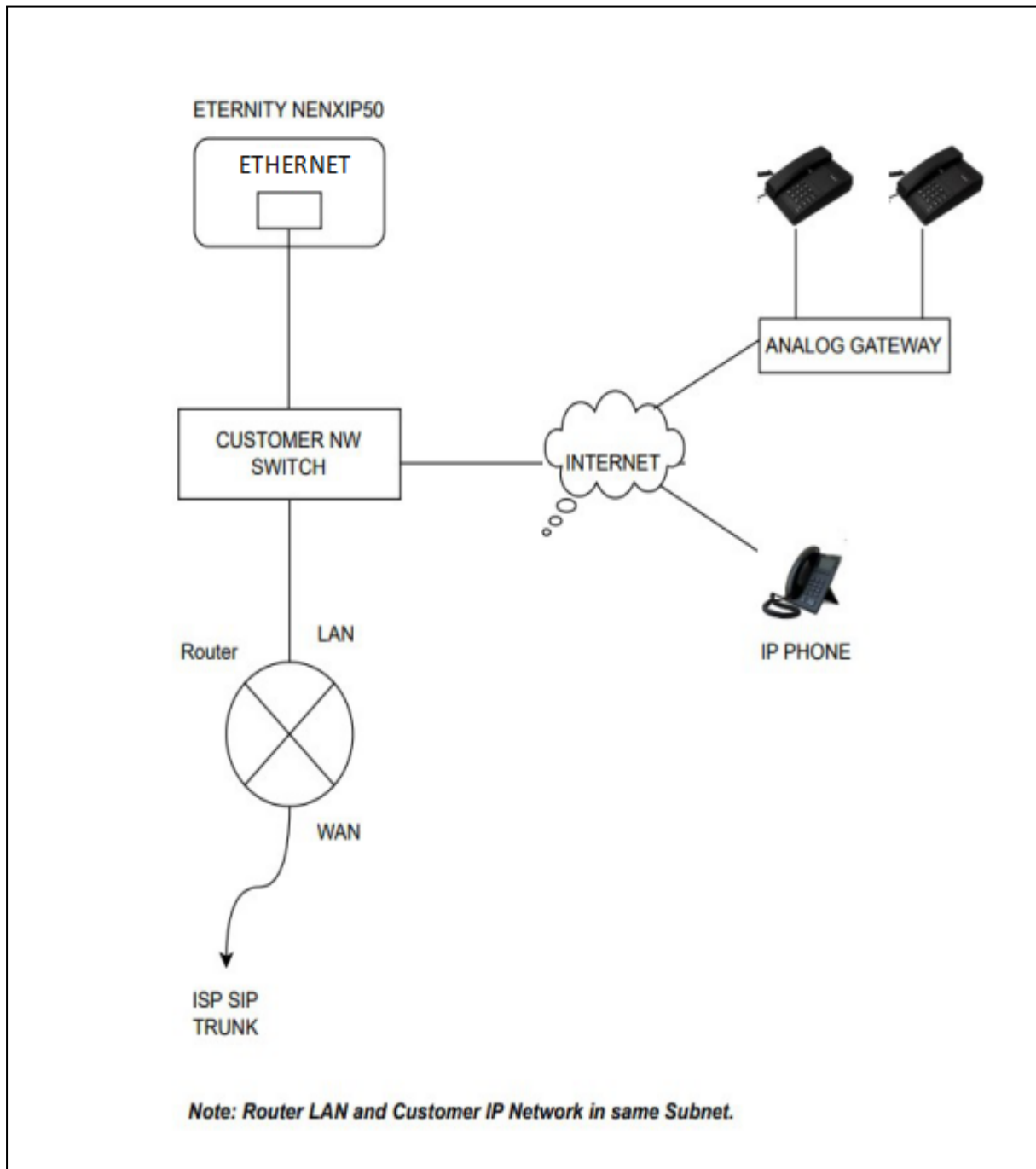
→ WebJeeves Users

System Details

Product Name	ETERNITY NE
Product Variant	ETERNITY NENXIP50
Firmware Version	V01R07.06.00
System Command	V01R01
Kernel Date	25-Mar-2022
IP Address	192.168.51.206
Master/CPU	V1R7.5.0
CPLD Version	V03R02
CO4 SLT2	V03R00
MOBILE1	V02R01
MOBILE1	V02R01
VoIP Server 8CH	V1R6.7.0
VMS4	V7R3.1.0
SMS Server	V3R1.4.0

Connectivity:

- Connect PC, Eternity NENX system and Router on the same switch.
- Connect ISP sip trunk on WAN Port of the Router.



Let us assume the SIP Trunk details given by JIO Service Provider as per below:

PILOT NUMBER	LOWER RANGE	HIGHER RANGE	CUSTOMER IP	GATEWAY IP	SBC IP	SUBNET MASK
+912653455111	+912653455100	+912653455130	10.64.74.182	10.64.74.181	100.64.216.4	255.255.255.0

Configuration Steps:

1. Router Configuration (Need to configure according to router)
2. JIO SIP Trunk Configuration on Network port
3. SIP Trunk Configuration
4. DDI Mapping
5. Outgoing Configuration

Step 1: Router Configuration - Scope of the partner/SI/Client (We are not suggesting any Router this is only just an example)

The screenshot displays the Tenda router's web interface for configuring an Internet connection. The main content area is titled "Internet Connection Setup" and features a dropdown menu set to "Static IP". Below this, several input fields are populated with the following values: IP Address (100.64.74.182), Subnet Mask (255.255.255.0), Gateway (100.64.74.181), DNS Server (8.8.8.8), and MTU (1500). An "Alternate DNS Server" field is present but empty, marked as optional. At the bottom of the form are "OK" and "Cancel" buttons. A sidebar on the left provides navigation options, with "Internet Connection Setup" currently selected. A help panel on the right explains that "Static IP" is a connection type where the user specifies a fixed IP address and other network parameters provided by their ISP.

192.168.1.190/advance.asp

Tenda

HomeAdvancedWirelessQoSApplicationsSecurityTools

Port Range Forwarding

DMZ Host

DDNS

UPNP Settings

Static Routing

Routing Table

DMZ Host

NOTE: When the DMZ host is enabled, the firewall settings of the DMZ host will not function.

DMZ Host IP Address192.168.1.101

☒ Enable

OKCancel

Help

The DMZ (De-Militarized Zone) function disables the firewall on the router for one device for a special service, such as Internet gaming or video conferencing.

DMZ Host IP Address:The IP address of the device for which the router's firewall will be disabled. Be sure to statically set the IP address of that device in the DHCP Client List Section to ensure that this function is consistent.

192.168.1.190

Tenda

HomeAdvancedWirelessQoSApplicationsSecurityTools

Status

Internet Connection Setup

MAC Clone

WAN Speed

LAN Settings

DNS Settings

DHCP Server

DHCP Client List

LAN Settings

This page is used to set the basic network parameters for LAN.

LAN MAC AddressC8:3A:35:43:96:60

IP Address192.168.1.190

Subnet Mask255.255.255.0

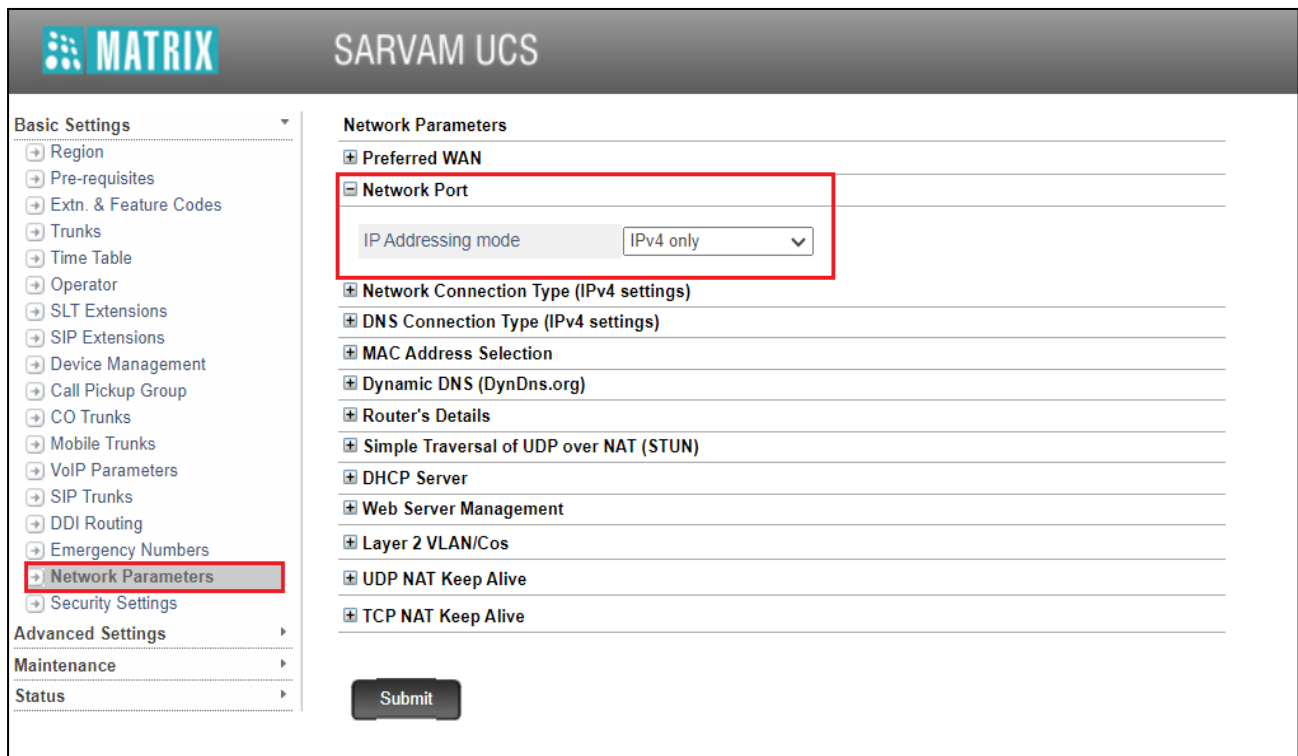
OKCancel

Help

Here you can set the LAN IP Address and Subnet Mask. This IP Address is to be used to access the router's homepage through a web browser. Be sure to make a note of any changes you apply to this page. Default settings are 192.168.0.1 and 255.255.255.0.

Step 2: JIO SIP Trunk Configuration on Network port

- Access Eternity NENX on **192.168.1.101** (Default IP) and Login into the page.
- Go to **Basic Settings** -> **Network Parameters** -> **Network Port** and Select **IPv4 only** as an **IP Addressing mode**.



MATRIX SARVAM UCS

Basic Settings

- Region
- Pre-requisites
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- Trunks
- Time Table
- Operator
- SLT Extensions
- SIP Extensions
- Device Management
- Call Pickup Group
- CO Trunks
- Mobile Trunks
- VoIP Parameters
- SIP Trunks
- DDI Routing
- Emergency Numbers
- Network Parameters**
- Security Settings

Advanced Settings

Maintenance

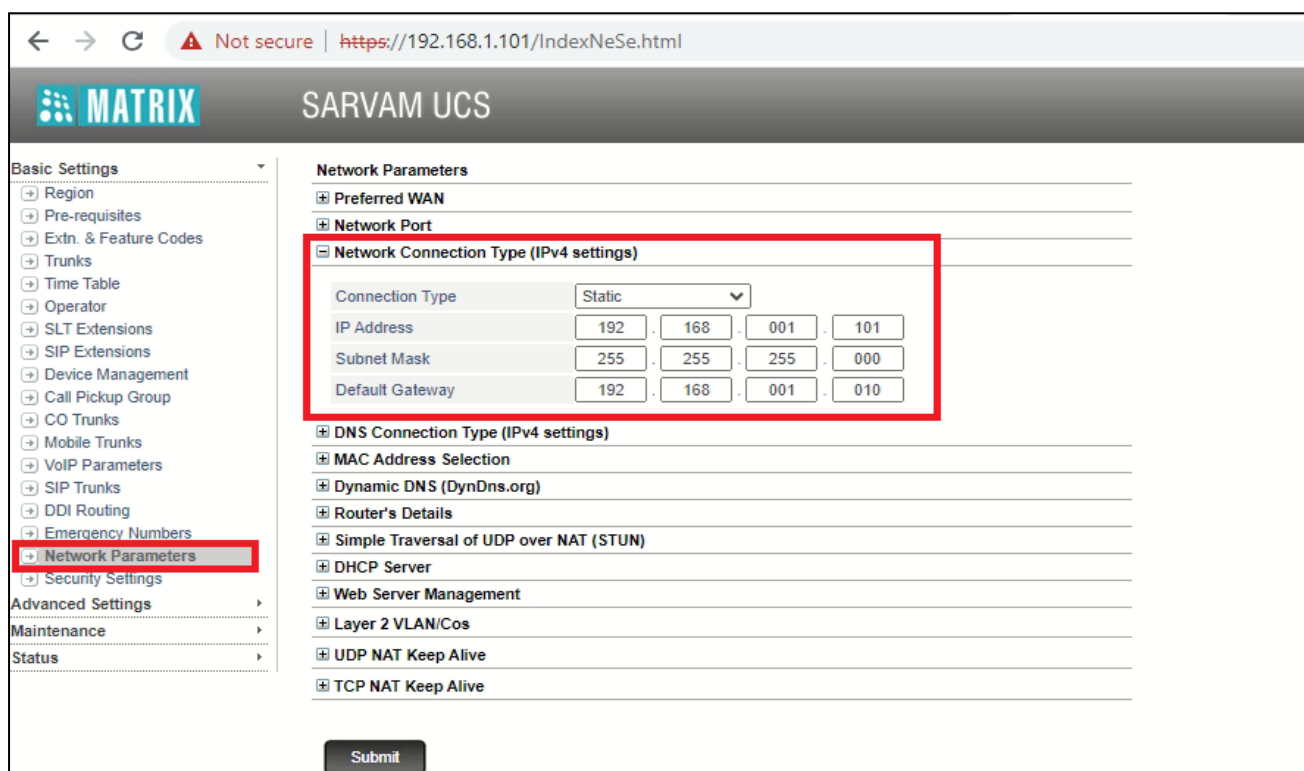
Status

Network Parameters

- Preferred WAN
- Network Port**
 - IP Addressing mode: IPv4 only
- Network Connection Type (IPv4 settings)
- DNS Connection Type (IPv4 settings)
- MAC Address Selection
- Dynamic DNS (DynDns.org)
- Router's Details
- Simple Traversal of UDP over NAT (STUN)
- DHCP Server
- Web Server Management
- Layer 2 VLAN/Cos
- UDP NAT Keep Alive
- TCP NAT Keep Alive

Submit

- In Network Connection Type, you can Configure **Eternity NENX** system IP address.



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- Emergency Numbers
- Network Parameters**
- Security Settings

Advanced Settings

Maintenance

Status

Network Parameters

- Preferred WAN
- Network Port
- Network Connection Type (IPv4 settings)**
 - Connection Type: Static
 - IP Address: 192 . 168 . 001 . 101
 - Subnet Mask: 255 . 255 . 255 . 000
 - Default Gateway: 192 . 168 . 001 . 010
- DNS Connection Type (IPv4 settings)
- MAC Address Selection
- Dynamic DNS (DynDns.org)
- Router's Details
- Simple Traversal of UDP over NAT (STUN)
- DHCP Server
- Web Server Management
- Layer 2 VLAN/Cos
- UDP NAT Keep Alive
- TCP NAT Keep Alive

Submit

- Go to the **Network Parameter -> Static Routing Table** and configure it as shown below.
- Destination Address: 100.64.216.0 (Server IP's last segment will be 000)
- Subnet Mask: 255.255.255.0
- Gateway Address: 192.168.1.190 (Router LAN IP address)

- Least Cost Routing
- Closed User Group
- COSEC Integration
- CTI
- Date & Time
- Default System
- Department Groups
- Dial Plan for SIP Extension
- DISA - CLI Authentication
- Emergency
- Key Template
- LDAP
- Least Cost Routing (LCR)
- License Management
- Logical Partition
- Macros
- Mobile Trunks
- Network Parameters
 - Static Routing Table**
 - Toll Control - Allowed-Denied Numbers
 - Page Zones
- Password
- PIN Configuration
- Regional Settings
- Response Mapping
- SMDR
- SMS Gateway
- SMS Routing
- SMS Server
- SMTP Settings
- System Log
- System Parameters
- System Timers and Counts

Static Routing Table

IPv4 Addresses

Index	Destination Address	Subnet Mask	Gateway Address
1	100 . 64 . 216 . 000	255 . 255 . 255 . 0	192 . 168 . 1 . 190
2	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
3	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
4	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
5	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
6	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
7	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
8	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000

IPv6 Addresses

Index	Destination Address	Prefix Length	Gateway Address
1		064	
2		064	
3		064	
4		064	
5		064	
6		064	
7		064	
8		064	

Note: Changing any entry on this page may affect ongoing VoIP calls associated with that entry.

- Also, configure the Destination Address: 10.64.217..0 (Media IP's last segment will be 000)
- Subnet Mask: 255.255.255.0
- Gateway Address: 192.168.1.190 (Router LAN IP address)

- Least Cost Routing
- Closed User Group
- COSEC Integration
- CTI
- Date & Time
- Default System
- Department Groups
- Dial Plan for SIP Extension
- DISA - CLI Authentication
- Emergency
- Key Template
- LDAP
- Least Cost Routing (LCR)
- License Management
- Logical Partition
- Macros
- Mobile Trunks
- Network Parameters
 - Static Routing Table**
 - Toll Control - Allowed-Denied Numbers
 - Page Zones
- Password
- PIN Configuration
- Regional Settings
- Response Mapping
- SMDR
- SMS Gateway
- SMS Routing
- SMS Server
- SMTP Settings
- System Log
- System Parameters
- System Timers and Counts

Static Routing Table

IPv4 Addresses

Index	Destination Address	Subnet Mask	Gateway Address
1	100 . 64 . 216 . 000	255 . 255 . 255 . 0	192 . 168 . 1 . 190
2	100 . 64 . 217 . 000	255 . 255 . 255 . 000	192 . 168 . 1 . 190
3	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
4	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
5	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
6	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
7	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000
8	000 . 000 . 000 . 000	000 . 000 . 000 . 000	000 . 000 . 000 . 000

IPv6 Addresses

Index	Destination Address	Prefix Length	Gateway Address
1		064	
2		064	
3		064	
4		064	
5		064	
6		064	
7		064	
8		064	

Note: Changing any entry on this page may affect ongoing VoIP calls associated with that entry.

- Go to **Maintenance** -> **Network Diagnosis** and check the ping test of the **Server IP**.
- There should be getting ping from Server IP.

MATRIX SARVAM UCS

Basic Settings ▸
Advanced Settings ▸
Maintenance ▾
 ▸ Certificate Management
 ▸ Configuration Backup/Restore
 ▸ Default System
 ▸ Firmware Management
 ▸ Logs
 ▸ **Network Diagnosis**
 ▸ Network Drive
 ▸ PCAP Trace
 ▸ Restart System
 ▸ SIP Extension
 ▸ System Debug
 ▸ VMS Debug
 ▸ VoIP Debug
Status ▸

Network Diagnosis
 ▢ Network Diagnosis

D diagnostic Utility ☒ Ping ☐ Traceroute

IP Address/Domain Name 100.64.216.4

Ping Packet Size 0032

Ping Count 04

Ping Timeout (sec) 3

Start **Default**

Diagnostic Result:

Clear

Connectivity

Step 3: SIP Trunk Configuration

- Go to **Basic Settings** -> **SIP Trunks** and configure SIP Trunk as Proxy.
- **Enable SIP Trunk.**
- Enter **SIP ID** which is given by the Service Provider.
- Select **SIP Trunk Mode** as **Proxy** and Disable the flag of **Check SIP ID during incoming call**.
- Select **Treat Incoming Call as Trunk**.
- Enter **Registrar Server Address** and **Port** as Server IP and Port which are given by Service Provider.
- Enter **Authentication User ID** and **Authentication Password** as per given by Service Provider.

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Pre-requisites
Extn. & Feature Codes
Trunks
Time Table
Operator
SLT Extensions
SIP Extensions
Device Management
Call Pickup Group
CO Trunks
Mobile Trunks
VoIP Parameters
SIP Trunks
DDI Routing
Emergency Numbers
Network Parameters
Security Settings

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Maintenance

Status

SIP-1

SIP-2

SIP-3

SIP-4

SIP-5

SIP-6

SIP-7

SIP-8

SIP

Enable SIP Trunk

☒

SIP Trunk Mode

☒ Proxy
☐ Peer-To-Peer

SIP ID

Check SIP ID during incoming call

☐

Treat incoming call as

Proxy/Registrar Parameters

Proxy/Registrar Server Address:Port

:

Authentication ID

Authentication Password

Outbound Proxy

Enable Outbound Proxy

☐

Outbound Proxy Address: Port

:

Incoming Call Routing

VoIP

Class of Service

Toll Control

Select Trunks for Outgoing Calls

.....

Note:

If you have set the SIP Trunk Mode to Peer-to-Peer and Treat incoming Peer-to-Peer call as Station, you may also configure Class of Service, Toll Control and Select Trunks for Outgoing Calls.

Submit

Default

Copy

- Submit the page and then click on **Advance/Down Arrow** ()
- Select **SIP ID** as **Send CLI in FROM Field**.

Basic Settings

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SIP-1

SIP-2

SIP-3

SIP-4

SIP-5

SIP-6

SIP-7

SIP-8

Incoming Call Routing

RCOC (Return Call to Original Caller)

Automatic Number Translation (ANT)

SMDR Storage

DSS Key Interface

Call Cost Calculation

Call Budget

Call Back

Call Taping

Call Duration Control

DTMF Out Dial

Send CLI in FROM field

Send CLI Option

Fixed Number

Send Called Party Number in

P-Asserted-Identity/P-Preferred-Identity for Outgoing calls

P-Asserted-Identity/P-Preferred-Identity for Incoming calls

VoIP

Caller ID on Call Transfer

Class of Service

Toll Control

Select Trunks for Outgoing Calls

.....

Note:

If you have set the SIP Trunk Mode to Peer-to-Peer and Treat incoming Peer-to-Peer call as Station, you may also configure Class of Service, Toll Control and Select Trunks for Outgoing Calls.

Submit

Default

Copy

- Go to **VoIP** > enable the Send options as Heartbeat

The screenshot shows the MATRIX SARVAM UCS web interface. On the left is a sidebar menu with categories: Basic Settings, Advanced Settings, Maintenance, and Status. Under Basic Settings, 'SIP Trunks' is highlighted with a red box. The main content area shows settings for SIP-1. Under the 'Supplementary Services' section, the 'Send OPTIONS As HeartBeat' checkbox is checked and highlighted with a red box. Other settings include Source Port IP Address (Use Ethernet Port IP Address), No. of Simultaneous Calls (08), Handle rport (Force NAT), Use Symmetric RTP (unchecked), Accept RTP Packets from Random Port (checked), Default Transport for Outgoing Message (UDP), Send REGISTER Message (checked), Add rinstance in REGISTER (checked), Allow registration for any contact (unchecked), Allow OG Calls without Registration (unchecked), DNS Record Type (A/AAAA Record), HeartBeat Interval (030), Send Re-Invite over SIP Trunk on Hold (Disable), Delayed Offer (Disable), Fetch Called Party Number From (Request URI), Display Called Party Number as CLI (unchecked), On Connect Media Send (183 Session Progress), Answer Source Trunk on Receiving (Early Media), and Use "tel" URI type in (None).

SIP-1	SIP-2	SIP-3	SIP-4	SIP-5	SIP-6	SIP-7	SIP-8
Supplementary Services							
Source Port IP Address		Use Ethernet Port IP Address					
No. of Simultaneous Calls		08					
Handle rport		Force NAT					
Use Symmetric RTP		☐					
Accept RTP Packets from Random Port		☒					
Default Transport for Outgoing Message		UDP					
Send REGISTER Message		☒					
Add rinstance in REGISTER		☒					
Allow registration for any contact		☐					
Allow OG Calls without Registration		☐					
DNS Record Type		A/AAAA Record					
Send OPTIONS As HeartBeat		☒					
HeartBeat Interval		030					
Send Re-Invite over SIP Trunk on Hold		Disable					
Delayed Offer		Disable					
Fetch Called Party Number From		Request URI					
Display Called Party Number as CLI		☐					
On Connect Media Send		183 Session Progress					
Answer Source Trunk on Receiving		Early Media					
Use "tel" URI type in		None					

- Set parameters of “**P-Asserted-Identity/P-Preferred Identity for Outgoing Calls**” as shown below only if you want to do outgoing calls through DDI mapped numbers otherwise disable this flag.

MATRIX

SARVAM UCS

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- CO Trunks
- Mobile Trunks
- VoIP Parameters
- **SIP Trunks**
- DDI Routing
- Emergency Numbers
- Network Parameters
- Security Settings

Advanced Settings ▸

Maintenance ▸

Status ▸

SIP-1
SIP-2
SIP-3
SIP-4
SIP-5
SIP-6
SIP-7
SIP-8

✚ SIP

✚ Trusted IP Address/es

✚ Incoming Call Routing

✚ RCOG (Return Call to Original Caller)

✚ Automatic Number Translation (ANT)

✚ SMDR Storage

✚ DSS Key Interface

✚ Call Cost Calculation

✚ Call Budget

✚ Call Back

✚ Call Taping

✚ Call Duration Control

✚ DTMF Out Dial

✚ Send CLI in FROM field

✚ **P-Asserted-Identity/P-Preferred-Identity for Outgoing calls**

Add P-Asserted-Identity/P-Preferred-Identity ☒

Identity P-Preferred-Identity ▾

Caller ID Calling Party Wise ▾

Fixed Number

✚ P-Asserted-Identity/P-Preferred-Identity for Incoming calls

✚ VoIP

✚ Caller ID on Call Transfer

✚ Class of Service

✚ Toll Control

✚ Select Trunks for Outgoing Calls

✚

Note:
If you have set the SIP Trunk Mode to **Peer-to-Peer** and **Treat incoming Peer-to-Peer call as Station**, you may a

⬆

Submit
Default
Copy

- Now check the **SIP Trunk Status**.


MATRIX

SARVAM UCS

Basic Settings

Advanced Settings

Maintenance

Status

System Details

System Usage

System Performance

Mailbox Status

USB Status

CO

Mobile

SIP Trunk

SIP Extension

SMS Gateway

Network

DSS Status

WebJeeves Users

SIP-1

SIP-2

SIP-3

SIP-4

SIP-5

SIP-6

SIP-7

SIP-8

SIP-1

StatusActive

Registration Time

Registration Retry Count

Reason of Failure

Call Budget

Call Budget TypeNone

Consumed Amount (₹)

Budget Reset ModeDisable

Budget Reset Scheduled(Date)1

Reset Consumed Budget

Reset

Step 4: DDI Mapping

- Go to **Basic Settings** -> **DDI Routing** -> Click on **DDI Number** and configure as shown below.
- Keep the **“Allow Incoming DDI Calls”** and **“Send DDI Number as CLI”** flag **Enabled**.

Extensions	DDI Number For SIP Trunk								Allow Incoming DDI Calls	Send DDI Number as CLI
	SIP Trunk-1	SIP Trunk-2	SIP Trunk-3	SIP Trunk-4	SIP Trunk-5	SIP Trunk-6	SIP Trunk-7	SIP Trunk-8		
21	5101								☒	☒
22	5102								☒	☒
23	5103								☒	☒
24	5104								☒	☒
25	5105								☒	☒
26	5106								☒	☒
27	5107								☒	☒
28	5108								☒	☒
29	5109								☒	☒
30	5110								☒	☒
SIP Extn-9									☒	☒
SIP Extn-10									☒	☒
SIP Extn-11									☒	☒
SIP Extn-12									☒	☒

- Now Click on the **Pilot Number** and configure as shown below.
- Click **Submit**.

Trunks	Pilot Number
SIP-1	+912653455111
SIP-2	
SIP-3	
SIP-4	
SIP-5	
SIP-6	
SIP-7	
SIP-8	

Submit **Default**

Step 5: Outgoing Configuration

- Go to **Basic Settings** -> **SLT Extensions** -> **Toll Control** and configure as shown below

The screenshot shows the SARVAM UCS configuration interface. On the left, the 'Basic Settings' menu is expanded, and 'SLT Extensions' is highlighted. The main content area shows the 'Toll Control' settings for extension 21. The 'Toll Control' section is highlighted with a red box and contains the following settings:

Setting	Value
Calls allowed during Day	All Calls
Calls allowed during Night/Break	All Calls
Calls allowed for Lock Level 1	No Calls
Calls allowed for Lock Level 2	No Calls
Calls allowed for Lock Level 3	No Calls

Below the 'Toll Control' section, there are several other settings: 'Select Trunks for Outgoing Calls', 'Hardware Settings', 'DND Intercept Routing', 'SMDR Storage', 'Call Budget', 'Caller ID on Call Transfer', 'Walk-In/Out', 'Call Forward', and a '.....' option. At the bottom, there are 'Submit', 'Default', and 'Copy' buttons.

- Now go to **"Select Trunks for Outgoing Calls"** and **"Double-Click to select"** the trunk.

The screenshot shows the SARVAM UCS configuration interface. On the left, the 'Basic Settings' menu is expanded, and 'SLT Extensions' is highlighted. The main content area shows the 'Select Trunks for Outgoing Calls' settings for extension 21. The 'Select Trunks for Outgoing Calls' section is highlighted with a red box and contains the following settings:

Setting	Value
Trunks allowed for dialing '0'	Double-click to select... LCR OFF
Trunks allowed for dialing '5'	Double-click to select... LCR OFF

Below the 'Select Trunks for Outgoing Calls' section, there are several other settings: 'Hardware Settings', 'DND Intercept Routing', 'SMDR Storage', 'Call Budget', 'Caller ID on Call Transfer', 'Walk-In/Out', 'Call Forward', and a '.....' option. At the bottom, there are 'Submit', 'Default', and 'Copy' buttons.

- The Pop-up Window will open, configured as shown below and Click on **OK**.

CO-1
CO-2
CO-3
CO-4
MOBILE-1
MOBILE-2
SIP-1
SIP-2
SIP-3
SIP-4
SIP-5
SIP-6
SIP-7
SIP-8

Select >>

SIP-1

↑
↓

☒ Rotation

OK Cancel

To remove trunks, use the **Delete** button on your keyboard

- Now **submit** the page.

MATRIX SARVAM UCS

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Basic Settings

- Region
- Pre-requisites
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- SIP Extensions
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- CO Trunks
- Mobile Trunks
- VoIP Parameters
- SIP Trunks
- DDI Routing
- Emergency Numbers
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- Security Settings

Advanced Settings

Maintenance

Status

Select Trunks for Outgoing Calls

Trunks allowed for dialing '0' SIP-1, LCR Number based

Trunks allowed for dialing '5' Double-click to select... LCR OFF

Hardware Settings

- DND Intercept Routing
- SMDR Storage
- Call Budget
- Caller ID on Call Transfer
- Walk-In/Out
- Call Forward

Submit Default Copy

This completes the entire configuration on SARVAM UCS SOHO.

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