



Avaya Solution & Interoperability Test Lab

Application notes for Frontier Communication System with Avaya™ Communication Server 1000 Release 6.0

Abstract

These Application Notes describe a solution comprised of Avaya™ Communication Server 1000E Release 6.0 and Frontier Communication SIP Trunk Product. The Primary focus of testing is the system verification of SIP trunk interoperability which includes the call scenarios such as basic call, call forward (all calls, busy, no answer), call transfer (blind and consultative) and conference. Calls should be placed in both directions and should involve various set types

Information in these Application Notes has been obtained through DevConnect compliance testing and additional technical discussions. Testing was conducted via the DevConnect Program at the Avaya Solution and Interoperability Test Lab.

Introduction

This document provides a typical network deployment of Communication Server 1000 (CS1000) utilizing the Frontier Communication SIP Trunking product offering. This document should serve as general guideline only, since it is not possible to document every possible variation of configuration. Further information may be obtained from your Avaya support representative.

The CS1000E system is configured as a SIP gateway endpoint on the Frontier Communication network. The enterprise customer will require an additional signaling server for each SIP gateway that will be deployed as SIP trunking to the carrier. In the diagram shown below, the signaling server is shown as the onboard CPPM Cores option, but it can also be the outboard, rack-mounted 1U server.

The CS1000, in this configuration, does not use SIP Redirect or Proxy for Carrier SIP trunking, the SIP Virtual Gateway is simply provisioned with the SBC as the static SIP endpoint of the SIP Trunk.

Interoperability Compliance Testing

System verification testing of SIP Trunking between CS1000 Rel. 6.0 and Carrier switch includes

- General call processing between systems including:
 - Codec/ptime negotiation and transcoding (G.711 u-law and G.729 verification / 20ms)
 - Hold/Retrieve on both ends
 - CLID displayed
 - Ringback tone
 - Speech path
 - Dialing plan support
 - Advanced features (Call on Mute, Call Park, Call Waiting, use Feature Access Code)
 - Abandoned Call
- Call redirection verification: all supported methods (blind transfer, consultative transfer, call forward, and conference) including CLID. Call redirection is performed from both ends
- Fax G711 Pass Through (Fax T38 does not support on Frontier)
- DTMF on both direction
- SIP Transport UDP
- Thru dialing via PBX Call Pilot
- Voice Mail Server (hosted on Avaya system)
- Early Media Transmission
- Inter-office tandem Call

Caveats

- The Fax/Modem pass through feature provides a modem pass through allowed (MPTA) class of service (CLS) for an analog phone TN. MPTA CLS dedicates an analog phone TN to a modem or a Fax machine terminal. A connection that initiates from the dedicated TN, and/or calls that terminate at the dedicated TN through a Digital Signal Processor (DSP), use a G711 NO VAD codec on the Call Server. To ensure proper functioning of the MPTA CLS, the **Enable Modem/Fax pass through mode** check box must be selected in the Gateways section of Element Manager. This check box is selected by default in Element Manager.
- The packet interval for G.711 codec is set to 20 ms in MPT. The maximum speed supported for modem and fax is 33.6 Kb/s. This limit is imposed by the analogue line card. When MPTA CLS is configured on a TN, the T.38 protocol is no longer supported for that particular TN.

Dependencies

- . CS1000 R6.0 software and implementation of latest patches
- . Frontier Communications provides support to setup, configure, and troubleshoot on carrier switch for the duration of the testing.

Support

For technical support on Frontier Communication system, please contact Frontier technical support at:

- Toll Free: (800) 239 4430
- <http://www.frontierhelp.com/techsupport.cfm>

Reference Configuration

Figure 1 illustrates the test configuration used during the compliance testing event between the Communication Server 1000E and Frontier Communication System. This configuration is for a single Communication Server1000E deployment

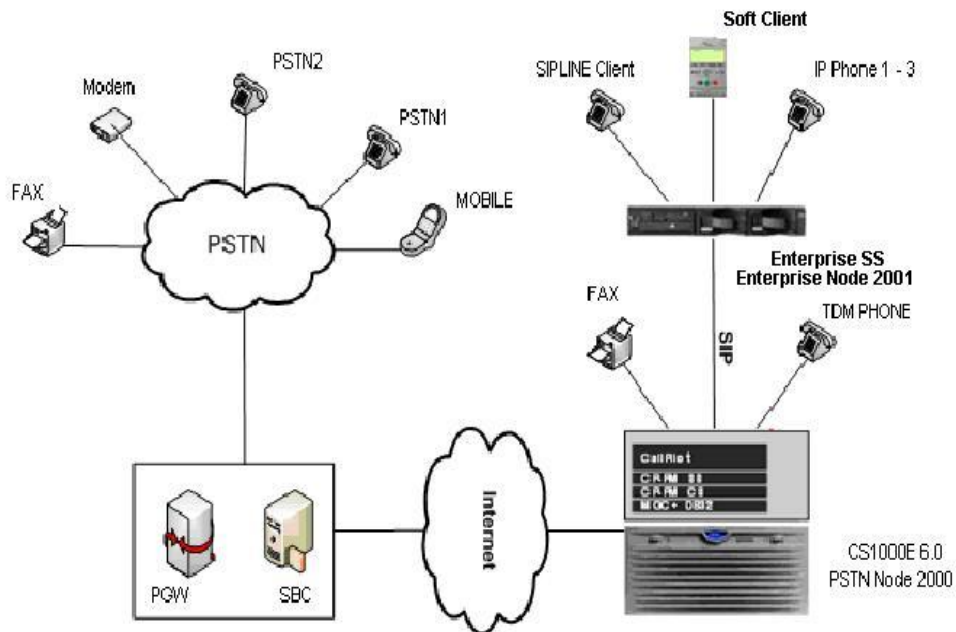


Figure 1- Network diagram for Avaya-Frontier LAB setup

Figure 2 depicts the deployment of two or more Communication Server of 1000E with the Frontier communication system.

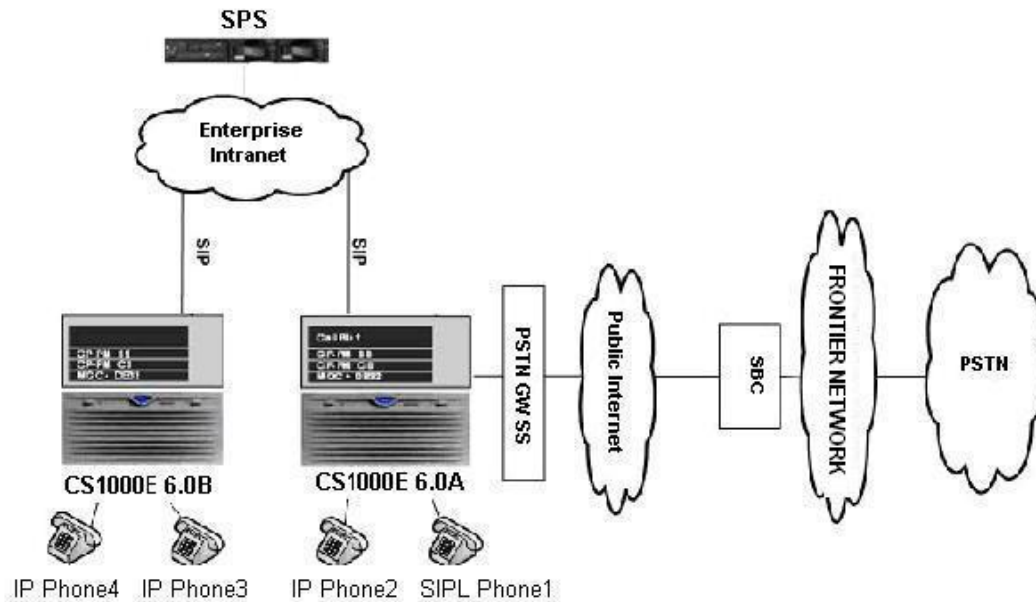


Figure 2 - Network topology for Multi-System configuration for Tandem Calls

The following assumptions were made for this lab test configuration:

1. CS1000 R6.0 software and implementation of latest patches
2. Frontier Communications provides support to setup, configure, and troubleshoot on carrier switch for the duration of the testing.

All test scenarios involving the establishment of calls will assume the following activities:

1. Calls will be checked for the correct call progress tones and cadences.
2. During the ringing state the ring back tone and destination ringing will be checked.
3. Calls will be checked in both hands-free and handset mode due to internal Avaya requirement.
4. Calls will be checked for speech path in both directions using spoken words to ensure clarity of speech.
5. The display(s) of the sets/clients involved will be checked for consistent and expected CLID, name and redirection information both prior to answer and after call establishment.
6. The speech path and messaging system will be observed for timely and quality End to End tone audio path generation and application responses.
7. The call server maintenance terminal window will be open during the test cases execution for the monitoring of BUG(s), ERR and AUD messages.
8. Speech path and display checked before and after calls are put on/off hold from each end.
9. Applicable of files will be screened on an hourly basis during the testing for message that may indicate technical issues. This refers to Avaya PBX files.

10. Calls will be checked to ensure that all resources such as Virtual trunks, TDM trunks, Sets and VGWs are released when a call scenario ends

Equipment and Software Validated

Additional software and patch lineup for the configuration is as follows:

Call Server: 6.00 with latest deplists loaded

Signaling Server: SSE 6.00.18 with latest DEPLIST

Patch ID	Issue	Title	Notes
MPLR28415	1	Ringback tone and speech path support in slow start CFNA scenarios	
MPLR28774	1	Delete element removes all elements-services mapping of associateroles	
MPLR28797	1	Unable to access overlays on inactive core when in split mode with UCM	
MPLR27408	1	SIP: Disable SIP Session Timer on CS1K.	
MPLR25946	1	SIP GW patch to remove outbound MCDN from SIP messaging	
MPLR22968	1	Replace domain population in the FROM field	
MPLR25529	1	PI: SIP: Partial support of DIVERSION	
MPLR27159	1	Mandatory parameter "T38FaxRateManagement" isn't present in T38 SDP	

Hardware system requirement and theirs software/loadware version

System	Software/Loadware Version
Avaya CS1000E 6.0 (CPPM)	<ul style="list-style-type: none"> ● Call Server: 6.00R ● Signaling Server: 6.00.18
Avaya phones	<ul style="list-style-type: none"> ● 2002 p2: 0604DCJ (Unistim) ● 2004 p2: 0604DCJ (Unistim) ● 1140: 0625C6O (Unistim) ● 1120: 0624C6O (Unistim) ● 2007: 0621C6M (Unistim) ● 1220: 062AC6O (Unistim) ● SIP 1140 i00v142 ● SIP 1120 ● SMC3456: Version 2.6 - RC14 build 53715
Genband C3	<ul style="list-style-type: none"> ● 7.2.40.40
Genband C6	<ul style="list-style-type: none"> ● 10.4.7

Configure the Avaya Communication Server 1000E

Element Manager Configuration

Configure IP in CS1000 network

This section describes the steps for creating Node ID (1000) in CS 1000 network. Enter Element Manager through the IE browser (in IE address bar, type IP address of the Node IP or TLAN of Signaling Server).

- Input Node ID and press Save
- Enter TLAN, ELAN IP addresses of Signaling Server.

Node 1000 was added to be configured as the SIP gateway to the carrier services.

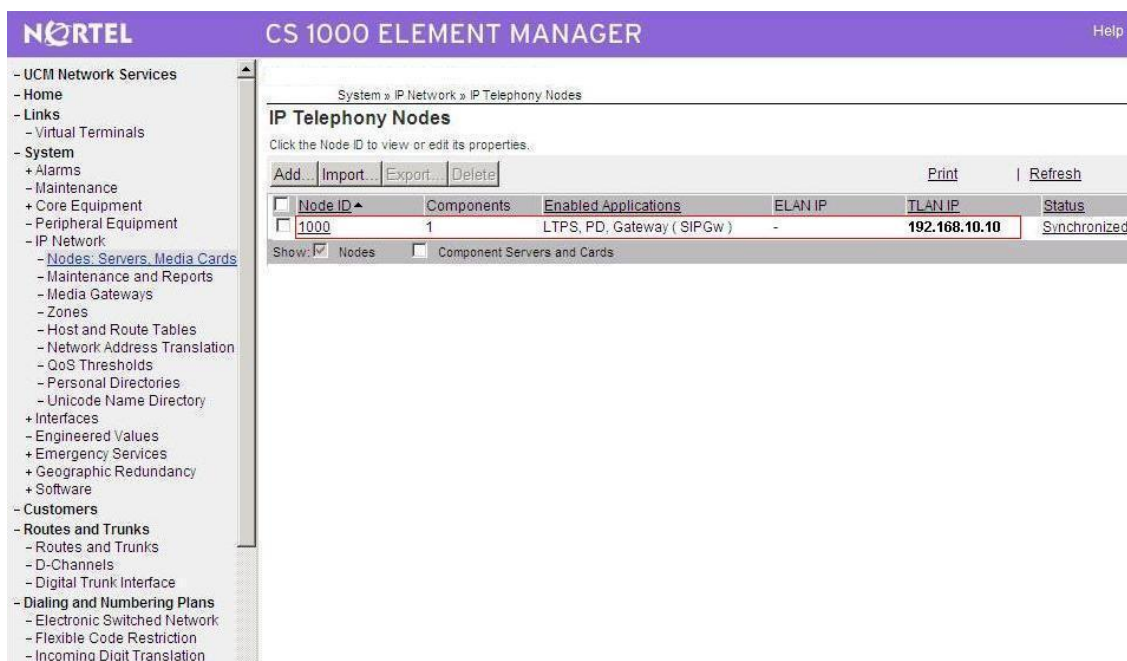


Figure 3 – Adding a node

Figure 4 describes the Call server IP configuration:

NORTEL CS 1000 ELEMENT MANAGER Help |

System » IP Network » IP Telephony Nodes » Node Details

Node Details (ID: 1000 - LTPS, PD, Gateway (SIPGw))

Node ID: (0-9999)

Call Server IP Address:

Telephony LAN (TLAN) Node IP Address: Embedded LAN (ELAN) Gateway IP address:

Subnet Mask: Subnet Mask:

IP Telephony Node Properties Applications (click to edit configuration)

- [Voice Gateway \(VGW\) and Codecs](#)
- [Quality of Service \(QoS\)](#)
- [Terminal Proxy Server \(TPS\)](#)

* Required Value.

Associated Signaling Servers & Cards

Select to add

Hostname	Type	Deployed Applications	ELAN IP	TLAN IP	Role
<input type="checkbox"/> nd1-car1	Signaling Server	LTPS, Gateway, PD	192.168.100.149	192.168.10.245	Leader

Note: Only server(s) that are not part of any other IP telephony node and deployed application(s) that match the service(s) selected for this node are available in the servers list.

Figure 4 - Call Server IP Configuration

Figure 5 – Virtual Trunk Gateway configuration

Configure Voice Codec for Avaya IP Phone

This section describes the steps for administering a set of codecs in CS1000. This set of codecs is used in IP network for communication between Avaya IP Phones.

- Access EM by IE browser.
- Choose **"IP Network"**, then choose **"Nodes: Servers, Media Cards"**, select proper **Node** and press **"Edit"**.

Figure 6 and 7 are showing how to change Codec profile for IP Phone by selecting "VGW and IP phone codec profile".

Disable Modem FAX pass through mode for G711 and check V.21 Fax tone Detection for tone detection by default. TN of sets with class of service =MPTD (Modem Pass Through Denied)

Figure 6 – Voice Gateway and Codec settings

CS 1000 ELEMENT MANAGER

Help

- Maintenance and Reports
 - Media Gateways
 - Zones
 - Host and Route Tables
 - Network Address Translation
 - QoS Thresholds
 - Personal Directories
 - Unicode Name Directory
- + Interfaces
 - Engineered Values
 - + Emergency Services
 - + Geographic Redundancy
 - + Software
- Customers
- Routes and Trunks
 - Routes and Trunks
 - D-Channels
 - Digital Trunk Interface
- Dialing and Numbering Plans
 - Electronic Switched Network
 - Flexible Code Restriction
 - Incoming Digit Translation
- Phones
 - Templates
 - Reports
 - Properties
 - Migration
- Tools
 - + Backup and Restore
 - Call Server Initialization
 - Date and Time
 - Logs and reports
- Security
 - + Passwords
 - + Policies
 - + Login Options

Managing: Username:

System » IP Network » IP Telephony Nodes » Node Details » VGW and Codecs

Node ID: 1000 - Voice Gateway (VGW) and Codecs

General | **Voice Codecs** | Fax

Echo Cancellation: ☒ Use canceller, with tail delay: 128

☒ Dynamic attenuation

Voice Activity Detection Threshold: -17 (-20 - +10 DBM)

Idle Noise Level: -65 (-327 - +327 DBM)

Signaling Options: ☒ DTMF Tone Detection

- ☐ Low latency mode
- ☒ Remove DTMF delay (squelch DTMF from TDM to IP)
- ☐ Modem/Fax pass-through
- ☒ V.21 Fax Tone Detection

Voice Codecs

Codec G711: ☒ Enabled (required)

Voice payload size: 20 (milliseconds per frame)

Voice Playout (jitter buffer) delay: 40 80 (milliseconds)

Nominal Maximum

Maximum delay may be automatically adjusted based on Nominal settings.

☐ Voice activity detection (VAD)

* Required Value.

Note: Changes made on this page will NOT be transmitted until the Node is also saved.

Save Can

Figure 7 shows how to configure the Voice gateway and IP phone codec settings. The Frontier Communication network supports both G.711 and G.729. The packet size is set to 20 to match the network also.

NORTEL CS 1000 ELEMENT MANAGER Help | Logout

Managing: 192.168.10.5 Username: admin

System » IP Network » IP Telephony Nodes » Node Details » VGW and Codecs

Node ID: 1000 - Voice Gateway (VGW) and Codecs

General | **Voice Codecs** | Fax

☐ Voice activity detection (VAD)

Codec G729: ☒ Enabled

Voice payload size: 20 (milliseconds per frame)

Voice Playback (jitter buffer) delay: 40 80 (milliseconds)

Nominal Maximum

Maximum delay may be automatically adjusted based on Nominal settings.

☐ Voice activity detection (VAD)

Codec G723.1: ☐ Enabled

Voice payload size: 30 (milliseconds per frame)

Voice Playback (jitter buffer) delay: 60 120 (milliseconds)

Nominal Maximum

Maximum delay may be automatically adjusted based on Nominal settings.

Coding rate: 5.3 (kbps)

Fax

Codec name: T.38 FAX

Maximum rate: 14400 (bps)

Fax TCF method: 2

Fax Playback Nominal Delay: 100 (0 - 300 milliseconds)

FAX No Activity Timeout: 20 (10 - 32000 milliseconds)

Packet size: 30 (bps)

* Required Value. Note: Changes made on this page will NOT be transmitted until the Node is also saved. **Save** **Cancel**

Figure 7 – Voice Gateway and codec settings

Configure Voice Codec for Media Gateways

This section describes the steps for administering a set of codecs in CS1000. This set of codec is used in IP network for communication through Media gateways.

- Access EM by IE browser.
- Choose "IP Network", then choose "Media gateways", select proper voice gateways
- To change Codec profile for IP Phone, select "VGW and IP phone codec profile".

Figure 8 shows how to configure the Voice Gateway and IP phone codec profile

Disable Modem FAX pass through mode

TN of sets with class of service = MPTD (Modem Pass Through Denied)

Voice gateway and IP phone codec settings.

NORTEL CS 1000 ELEMENT MANAGER

Help

– UCM Network Services

– Home

– Links

– Virtual Terminals

– System

– Alarms

– Maintenance

– Core Equipment

– Peripheral Equipment

– IP Network

– Nodes: Servers, Media Cards

– Maintenance and Reports

– Media Gateways

– Zones

– Host and Route Tables

– Network Address Translation

– QoS Thresholds

– Personal Directories

– Unicode Name Directory

– Interfaces

– Application Module Link

– Value Added Server

– Property Management System

– Engineered Values

– Emergency Services

– Geographic Redundancy

– Software

– Customers

– Routes and Trunks

– Routes and Trunks

– D-Channels

– Digital Trunk Interface

– Dialing and Numbering Plans

– Electronic Switched Network

– VGW and IP phone codec profile

Enable echo canceller ☒

Echo canceller tail delay (milliseconds)

Enable dynamic attenuation ☒

Voice activity detection threshold (0 - 4 DBM)

Idle noise level (0 - 1 DBM)

DTMF tone detection ☒

Enable low latency mode ☐

Remove DTMF delay (squelch DTMF from TDM to IP) ☒

Enable modem/fax pass through mode ☐

Enable V.21 FAX tone detection ☒

Fax TCF method

FAX maximum rate (bps)

FAX playout nominal delay (0 - 300 milliseconds)

FAX no activity timeout (10 - 32000 milliseconds)

FAX packet size

+ Codec G711 Select ☒

+ Codec G729A Select ☒

+ Codec G723.1 Select ☒

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Figure 8 – Voice Gateway and IP phone codec profile settings

Configure Quality of Service

This section describes the steps for administering QoS in CS1000.

- Access EM by IE browser
- Choose "IP Network", then choose "Nodes: Servers, Media Cards", select proper Node and press "Edit".
- To change Quality of Service, select "
- QoS".

The default Diffserv values are correct in figure 9.

NORTEL CS 1000 ELEMENT MANAGER

Help

– UCM Network Services

– Home

– Links

– Virtual Terminals

– System

– Alarms

– Maintenance

– Core Equipment

– Peripheral Equipment

– IP Network

– Nodes: Servers, Media Cards

– Maintenance and Reports

– Media Gateways

– Zones

– Host and Route Tables

– Network Address Translation

– QoS Thresholds

– Personal Directories

– Unicode Name Directory

– Interfaces

– Application Module Link

– Value Added Server

– Property Management System

– Engineered Values

– Emergency Services

– Geographic Redundancy

– Software

– Customers

– Routes and Trunks

– Routes and Trunks

– D-Channels

– Digital Trunk Interface

– Dialing and Numbering Plans

– Electronic Switched Network

System > IP Network > IP Telephony > Nodes > Node Details > Quality of Service (QoS)

Node ID: 1000 - Quality of Service (QoS)

Diffserv Codepoint (DSCP)

Enable Nortel Automatic QoS ☐

Control Packets (0-63)

Voice Packets (0-63)

VLAN Tagging: ☐ 802.1Q Support

802.1Q Bits Value (802.1P) (0-7)

* Required Value.

Note: Changes made on this page will NOT be transmitted until the Node is also saved.

Save Cancel

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Figure 9 – Quality of Service settings.

Configure SIP URI

This section describes the steps for administering SIP URI configuration in CS1000.

- Access EM by IE browser
- Choose "IP Network", then choose "Nodes: Servers, Media Cards", select proper Node and press "Edit".
- To change SIP URI, select "SIP URI Map".

In figure 10, leave the SIP URI fields blank for E.164.

The screenshot shows the Nortel CS 1000 Element Manager interface. The top bar displays the Nortel logo and the title 'CS 1000 ELEMENT MANAGER'. Below the bar, the left sidebar contains a tree view of the system configuration, with 'Nodes: Servers, Media Cards' selected. The main content area shows the 'Node ID: 1000 - Virtual Trunk Gateway Configuration Details' page. The 'SIP Gateway Services' tab is active, displaying various configuration fields. The 'SIP URI Map' section includes fields for National, Subscriber, Special number, and Unknown, each with a dropdown menu. The 'SIP Gateway Services' section includes a checkbox for 'SIP Converged Desktop', a text field for 'Service DN', a text field for 'Converged telephone call forward DN', a text field for 'RAN route for Announce', a text field for 'Wait time before RAN queue', a text field for 'Timeout for ringing indication', and a text field for 'Timeout for CD server'. The 'SIP Gateway Services' section also includes a 'Note: Changes made on this page will NOT be transmitted until the Node is also saved.' and 'Save' and 'Cancel' buttons.

Figure 10 – SIP Gateway Services Settings

Configure Zones and Bandwidth Management

This section describes the steps for administering Zone configuration in CS1000.

- Access EM by IE browser
- Choose "IP Network", then choose "Zones", select proper "Zone Basic Property and Bandwidth Management"

Figure 11 shows how to configure a zone for IP sets and bandwidth management. If it does not exist already, create a zone for IP sets. The bandwidth strategy can be adjusted to preference.

Figure 12 shows how to configure a zone for newly created SIP trunks.

NORTEL CS 1000 ELEMENT MANAGER Help |

System » IP Network » Zones » Bandwidth Zones » Bandwidth Zones 10 » Zone Basic Property and Bandwidth Management

Zone Basic Property and Bandwidth Management

Input Description	Input Value
Zone Number (ZONE):	10
Intrazone Bandwidth (INTRA_BW):	100000
Intrazone Strategy (INTRA_STGY):	Best Quality (BQ)
Interzone Bandwidth (INTER_BW):	100000
Interzone Strategy (INTER_STGY):	Best Quality (BQ)
Resource Type (RES_TYPE):	Shared (SHARED)
Zone Intent (ZBRN):	MO (MO)
Description (ZDES):	

Submit Refresh Delete Cancel

Figure 11 – Zone Basic Property Setting for IP phones

NORTEL CS 1000 ELEMENT MANAGER Help |

System » IP Network » Zones » Bandwidth Zones » Bandwidth Zones 255 » Zone Basic Property and Bandwidth Management

Zone Basic Property and Bandwidth Management

Input Description	Input Value
Zone Number (ZONE):	255
Intrazone Bandwidth (INTRA_BW):	1000000
Intrazone Strategy (INTRA_STGY):	Best Quality (BQ)
Interzone Bandwidth (INTER_BW):	1000000
Interzone Strategy (INTER_STGY):	Best Quality (BQ)
Resource Type (RES_TYPE):	Shared (SHARED)
Zone Intent (ZBRN):	VTRK (VTRK)
Description (ZDES):	

Submit Refresh Delete Cancel

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Figure 12 – Zone Basic Property Settings for (virtual) SIP trunk

Configure SIP trunk

This section describes the steps for establishing a SIP connection between CS 1000 switch and Carrier system.

1. Create D-channel (DCH)

- Launch Element Manager of CS 1000 6.0
- Choose **D-Channels**, enter D-channel number (i.e.: 100), select DCH for type

Click **Add** to create DCH 100

Nortel CS 1000 ELEMENT MANAGER

Managing: **192.168.10.5**
Routes and Trunks » **D-Channels** » D-Channels 100 Property Configuration

D-Channels 100 Property Configuration

- Basic Configuration

Input Description	Input Value
Action Device And Number (ADAN) (TYPE)	DCH
D channel Card Type (CTYP)	DCHP
Designator (DES)	VoIP
Recovery to Primary (RCVP)	<input type="checkbox"/>
PRI loop number for Backup D-channel (BCHL)	
User (USR)	Integrated Services Signaling Link Dedicated (ISLD)
Interface type for D-channel (IFC)	Meridian Meridian1 (SL1)
Country (CNTY)	ETS 300=102 basic protocol (ETSI)
D-Channel PRI loop number (DCHL)	
Primary Rate Interface (PRI)	<input type="text"/> more PRI
Secondary PRI2 loops (PRI2)	<input type="text"/>
Meridian 1 node type (SIDE)	Slave to the controller (USR)
Release ID of the switch at the far end (RLS)	6
Central Office switch type (CO_TYPE)	100% compatible with Bellcore standard (STD)
Integrated Services Signaling Link Maximum (ISLM)	4000 Range: 1 - 4000
Signaling Server Resource Capacity (SSRC)	1800 Range: 0 - 4000

+ Basic options (BSCOPT)
+ Advanced options (ADVOPT)

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Figure 13 – D-Chanel Configurations

Also click on Basic Options and edit the Remote Capabilities (RCAP). Enable **MWI** if CS1K hosted voice mail will be used.

2. Create route: Create route 100 using DCH 100 for SIP trunks with figures 14 and 15

Managing: 192.168.10.5
Routes and Trunks » Routes and Trunks » Customer 0, Route 100 Property Configuration

Customer 0, Route 100 Property Configuration

- Basic Configuration

Input Description	Input Value
Route Data Block (RDB) (TYPE)	RDB
Customer number (CUST)	00
Route Number (ROUT)	100
Designator field for trunk (DES)	CARRIER
Trunk Type (TKTP)	TIE
Incoming and Outgoing trunk (ICOG)	Incoming and Outgoing (IAO)
Access Code for the trunk route (ACOD)	8100
Trunk type M911P (M911P)	<input type="checkbox"/>
The route is for a virtual trunk route (VTRK)	<input checked="" type="checkbox"/>
Zone for codec selection and bandwidth management (ZONE)	255 Range: 0 - 255
Node ID of signaling server of this route (NODE)	2000 Range: 0 - 9999
Protocol ID for the route (PCID)	SIP (SIP)
Print Correlation ID in CDR for the route (CRID)	<input type="checkbox"/>
Integrated Services Digital Network option (ISDN)	<input checked="" type="checkbox"/>
Mode of operation (MODE)	Route uses ISDN Signaling Link (ISLD)
D channel number (DCH)	100 Range: 0 - 254
Interface type for route (IFC)	Meridian M1 (SL1)

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Figure 14 – Route Property Configuration

- D channel number (DCH) 100 Range: 0 - 254

- Interface type for route (IFC) Meridian M1 (SL1)

- Private network identifier (PNI) 00001 Range: 0 - 32700

- Network calling name allowed (NCNA) ☒

- Network call redirection (NCRD) ☒

- Trunk route optimization (TRO) ☐

- Recognition of DTI2 ABCD FALT signal for ISL (FALT) ☐

- Channel type (CHTY) B-channel (BCH)

- Call type for outgoing direct dialed TIE route (CTYP) Unknown Call type (UKWN)

- Insert ESN access code (INAC) ☒

- Integrated service access route (ISAR) ☐

- Display of access prefix on CLID (DAPC) ☐

- Basic Route Options

Attendant announcement (ATAN) No Attendant Announcement (NO)

Billing number required (BILN) ☐

Call detail recording (CDR) ☐

North American toll scheme (NATL) ☒

Controls or timers (CNTL) ☐

Conventional (Tie trunk only) (CNVT) ☐

Incoming DID digit conversion on this route (IDC) ☒

Day IDC tree number (DCNO) 1 Range: 0 - 254

Night IDC tree number (NDNO) 1 Range: 0 - 254

Display external dialed digits (DEXT) ☐

Multifrequency compelled or MFC signaling (MFC) No MFC (NO)

Process notification networked calls (PNNC) ☐

Figure 15 – Route Property Configuration (Cont..)

3. Create trunk: Create trunk using basic configuration in figure 16

The screenshot displays the Nortel CS 1000 Element Manager interface. The left sidebar contains a navigation menu with categories: Home, Links, System, Customers, Routes and Trunks, Dialing and Numbering Plans, Tools, and Security. The 'Routes and Trunks' section is expanded, showing 'Routes and Trunks' as the selected item. The main content area is titled 'Customer 0, Route 100, Trunk 1 Property Configuration'. It features a 'Basic Configuration' section with a table of input fields. The table has two columns: 'Input Description' and 'Input Value'. The fields include: Trunk data block (TYPE) with value IPTI; Terminal Number (TN) with value 100 0 00 00; Designator field for trunk (DES) with value VOIP; Extended Trunk (XTRK) with value VTRK; Route number, Member number (RTMB) with value 100 1; Level 3 Signaling (SIGL) with a dropdown arrow; Card Density (CDEN) with value 8D; Start arrangement Incoming (STR) with value Immediate (IMM); Start arrangement Outgoing (STRO) with value Immediate (IMM); Trunk Group Access Restriction (TGAR) with value 0; Channel ID for this trunk, (CHID) with value 1; Increase or decrease the member numbers (INC) with value Increase channel and member number (YES); and Class of Service (CLS) with value Edit. Below the table is an 'Advanced Trunk Configurations' section, which is currently empty. At the bottom right of the configuration area are three buttons: Save, Delete, and Cancel. The footer of the interface shows the copyright notice: Copyright © 2002-2007 Nortel Networks. All rights reserved.

Input Description	Input Value
Trunk data block (TYPE)	IPTI
Terminal Number (TN)	100 0 00 00
Designator field for trunk (DES)	VOIP
Extended Trunk (XTRK)	VTRK
Route number, Member number (RTMB)	100 1
Level 3 Signaling (SIGL)	
Card Density (CDEN)	8D
Start arrangement Incoming (STR)	Immediate (IMM)
Start arrangement Outgoing (STRO)	Immediate (IMM)
Trunk Group Access Restriction (TGAR)	0
Channel ID for this trunk, (CHID)	1
Increase or decrease the member numbers (INC)	Increase channel and member number (YES)
Class of Service (CLS)	Edit

Figure 16 – Basic Trunk Configuration

Disable Media Security (sRTP) at the trunk level using figure 17 by editing the Class of Service (CLS) at the bottom basic trunk configuration page shown in figure 17

Nortel CS 1000 ELEMENT MANAGER Help | Logout

Class of Service Configuration

Input Description	Input Value
- ACD Priority (CLS)	ACD Priority not required (APN)
- Analog Semi-Permanent Connections (CLS)	Analog Semi-Permanent Connections Denied (SPCD)
- ARF Supervised COT (CLS)	
- Barring (CLS)	
- Battery Supervised COT (CLS)	
- Busy Tone Supervised COT (CLS)	
- Calling Line Identification (CLS)	
- Calling party (CLS)	Calling party Denied (CND)
- Central Office Ringback (CLS)	
- Centrex Switchhook Flash (CLS)	Centrex Switchhook Flash Denied (THFD)
- Dial Pulse (CLS)	Digitone (DTN)
- DTR PAD value (CLS)	
- Echo Canceling (CLS)	Echo Canceling Denied (ECD)
- Hong Kong DTI (CLS)	
- Loop Break Supervised COT (CLS)	
- Make-break ratio for dial pulse (CLS)	10 pulses per second (P10)
- Manual Incoming (CLS)	Manual Incoming Denied (MID)
- Media Security (CLS)	Media Security Never (MSNV)
- Network Hook Flash Over M911P (CLS)	
- Polarity (CLS)	

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Figure 17 – Class of Service

Since the carrier does not support TLS or sRTP, we have them disabled in our CS1K configuration. For the primary proxy, Enter the IP address of the Session Border Controller (SBC). Use UDP SIP transport, port 5060 for SIP communication. The NRS is not enabled as all calls are routed by the SBC.

4. Create Dialing Plan:

Create Special number list:

Create special number list for outgoing dialing plan using figure 18

Launch Element Manager of CS 1000 6.0

Select “Dialing and Numbering plans → Electronic Switched Network → Number Plan (Net) → Access Code 1 (2) → Special Number (SPN).

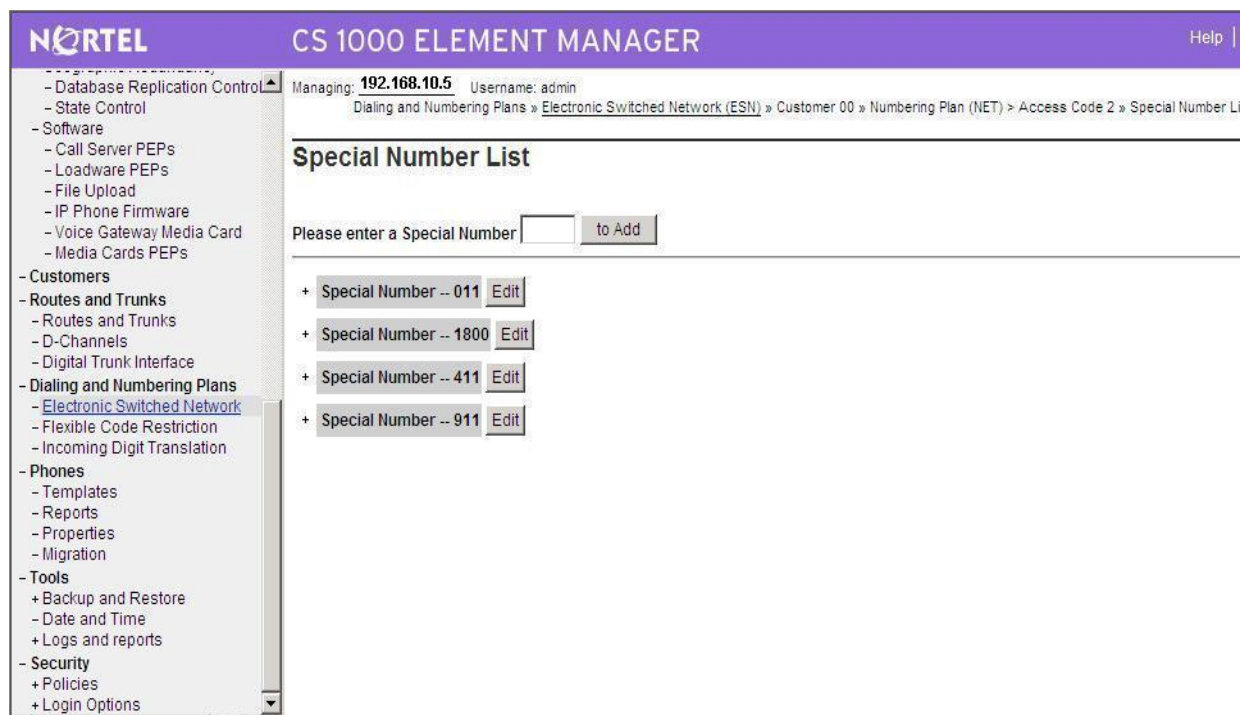


Figure 18 – Special Number List

Create special number SPN 011 (Use RLI_10) for outgoing dialing plan to International calls

The screenshot shows the Nortel CS 1000 Element Manager interface. The left sidebar contains a navigation menu with categories like Home, Links, System, Customers, Routes and Trunks, Dialing and Numbering Plans, Tools, and Security. The main content area is titled 'Special Number' and displays a configuration form. The form includes fields for 'Special Number translation (SPN)' set to '011', 'Flexible Length (FLEN)' set to '15', 'International Dialing Plan (INPL)' as an unchecked checkbox, 'Inhibit Time-out Handler (ITOH)' as an unchecked checkbox, 'Route List Index (RLI)' set to '10', and 'Type of call that is defined by the special number (CLTP)' set to 'No call type (NONE)'. There are also empty text areas for 'Number to be Denied (DENY)', 'Local DID number to be recognized (LDID)', and 'Local DDD number to be recognized (LDDD)'. The bottom of the page shows a copyright notice for 2002-2007 Nortel Networks.

Input Description	Input Value
Special Number translation (SPN):	011
Flexible Length (FLEN):	15 (0 - 24)
International Dialing Plan (INPL):	<input type="checkbox"/>
Inhibit Time-out Handler (ITOH):	<input type="checkbox"/>
Route List Index (RLI):	10
Type of call that is defined by the special number (CLTP):	No call type (NONE)
Number to be Denied (DENY): (Items separated by a space)	
Digit Manipulation Index for LDID Numbers (DMI):	1
Local DID number to be recognized (LDID): (Items separated by a space)	
Local DDD number to be recognized (LDDD): (Items separated by a space)	

Figure 19 – Special Number for International Calls

Create special number SPN 1800 (Use RLI_10) for outgoing dialing plan to toll free calls

The screenshot shows the Nortel CS 1000 Element Manager interface for configuring a special number. The left sidebar is identical to the previous figure. The main content area is titled 'Special Number' and displays a configuration form. The form includes fields for 'Special Number translation (SPN)' set to '1800', 'Flexible Length (FLEN)' set to '12', 'International Dialing Plan (INPL)' as an unchecked checkbox, 'Inhibit Time-out Handler (ITOH)' as an unchecked checkbox, 'Route List Index (RLI)' set to '10', and 'Type of call that is defined by the special number (CLTP)' set to 'No call type (NONE)'. There are also empty text areas for 'Number to be Denied (DENY)', 'Local DID number to be recognized (LDID)', and 'Local DDD number to be recognized (LDDD)'. The bottom of the page shows a copyright notice for 2002-2009 Nortel Networks.

Input Description	Input Value
Special Number translation (SPN):	1800
Flexible Length (FLEN):	12 (0 - 24)
International Dialing Plan (INPL):	<input type="checkbox"/>
Inhibit Time-out Handler (ITOH):	<input type="checkbox"/>
Route List Index (RLI):	10
Type of call that is defined by the special number (CLTP):	No call type (NONE)
Number to be Denied (DENY): (Items separated by a space)	
Digit Manipulation Index for LDID Numbers (DMI):	1
Local DID number to be recognized (LDID): (Items separated by a space)	
Local DDD number to be recognized (LDDD): (Items separated by a space)	

Figure 20 – Special Number for Tool Free Call

Create special number SPN 411 (Use RLI_10) for outgoing dialing plan to 411 service calls in figure 21

The screenshot shows the Nortel CS 1000 Element Manager interface. The left sidebar contains a navigation menu with categories: Home, Links, System, Customers, Routes and Trunks, Dialing and Numbering Plans, Tools, and Security. The 'Dialing and Numbering Plans' section is expanded, showing 'Electronic Switched Network' as the selected option. The main content area displays the 'Special Number' configuration page. At the top, it indicates the managed IP is 192.168.10.5 and shows the navigation path: Dialing and Numbering Plans > Electronic Switched Network (ESN) > Customer 00 > Numbering Plan (NET) > Access Code 1 > Special Number. The configuration form includes fields for 'Special Number translation (SPN)' set to 411, 'Flexible Length (FLEN)' set to 4, 'International Dialing Plan (INPL)' as an unchecked checkbox, 'Inhibit Time-out Handler (ITOH)' as an unchecked checkbox, 'Route List Index (RLI)' set to 10, and 'Type of call that is defined by the special number (CLTP)' set to 'No call type (NONE)'. There are also empty text areas for 'Number to be Denied (DENY)' and 'Local DID number to be recognized (LDID)', and a 'Digit Manipulation Index for LDID Numbers (DMI)' set to 1. The footer shows the copyright notice: Copyright © 2002-2007 Nortel Networks. All rights reserved.

Figure 21 – Special Number for 411 Service Call

Create special number 911 (use RLI_10) to dial to Emergency service in figure 22

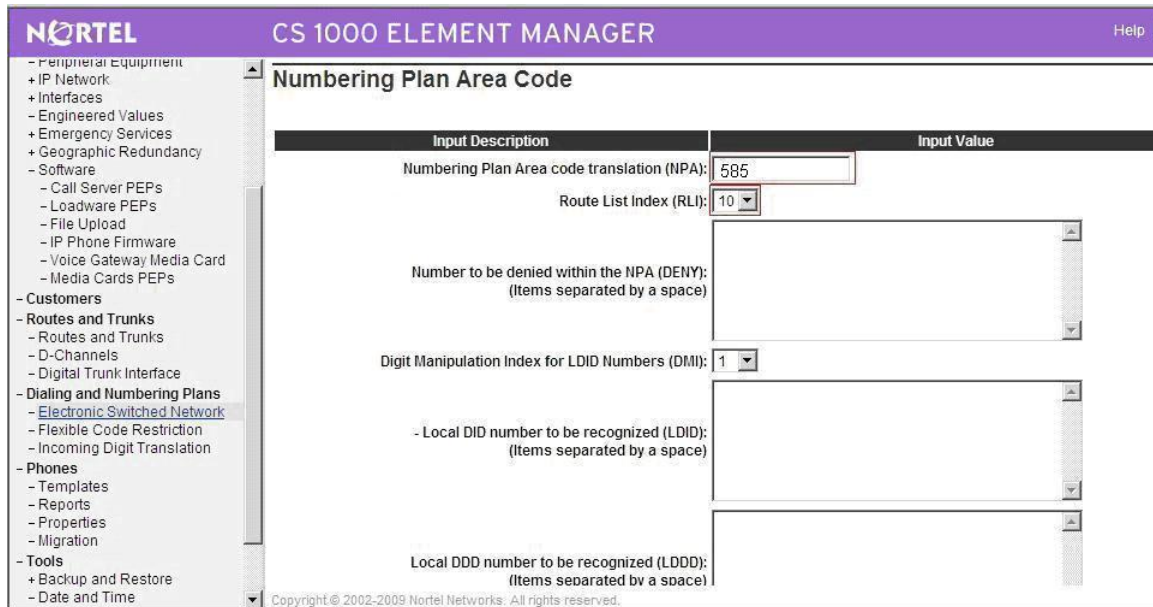
This screenshot is identical to Figure 21, showing the Nortel CS 1000 Element Manager interface. The configuration for the 'Special Number' is shown with 'Special Number translation (SPN)' set to 911. All other settings, including 'Flexible Length (FLEN)' at 4, 'Route List Index (RLI)' at 10, and 'Type of call that is defined by the special number (CLTP)' as 'No call type (NONE)', remain the same as in Figure 21. The interface elements, including the sidebar and footer, are also consistent with the previous figure.

Figure 22 – Special Number for Emergency 911 dialing

Create Numbering Plan Area Code:

Create NPA numbers for outgoing.

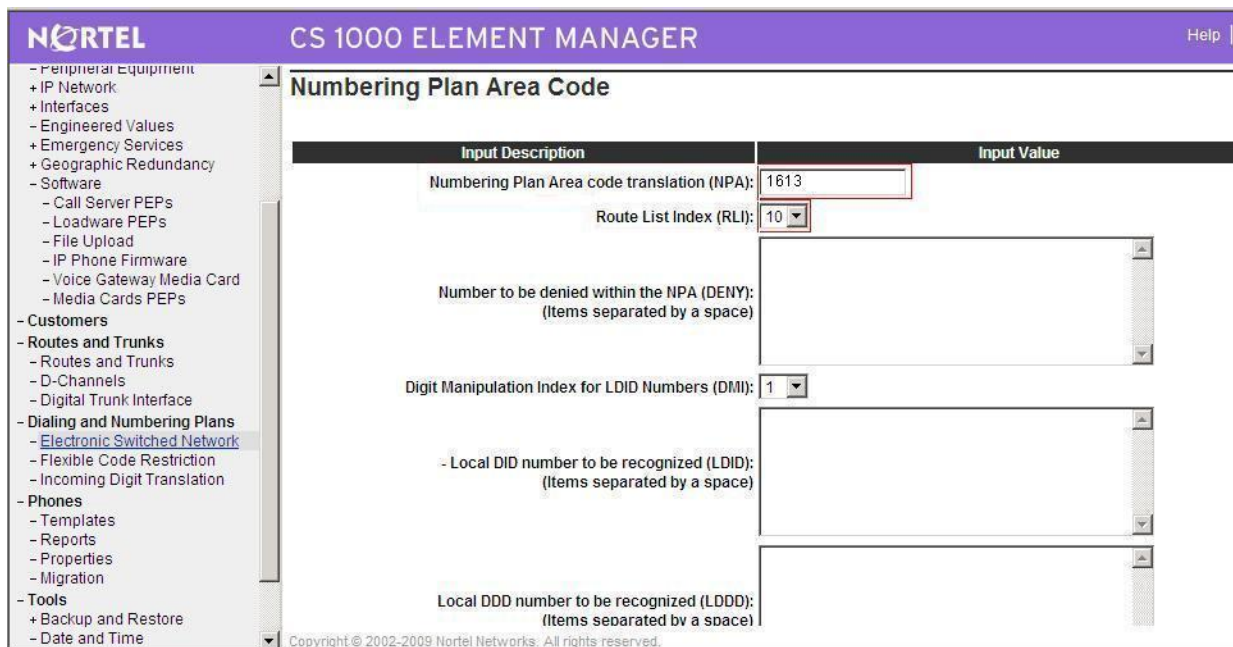
NPA_1585: Create NPA_1585 for outgoing calls to numbers beginning with 585; Figure 23



Input Description	Input Value
Numbering Plan Area code translation (NPA):	585
Route List Index (RLI):	10
Number to be denied within the NPA (DENY): (Items separated by a space)	
Digit Manipulation Index for LDID Numbers (DMI):	1
Local DID number to be recognized (LDID): (Items separated by a space)	
Local DDD number to be recognized (LDDD): (Items separated by a space)	

Figure 23 – Create NPA_1585 for outgoing calls

NPA_1613: Create NPA 1613 to dial to national DID numbers beginning with 613; Figure 24



Input Description	Input Value
Numbering Plan Area code translation (NPA):	1613
Route List Index (RLI):	10
Number to be denied within the NPA (DENY): (Items separated by a space)	
Digit Manipulation Index for LDID Numbers (DMI):	1
Local DID number to be recognized (LDID): (Items separated by a space)	
Local DDD number to be recognized (LDDD): (Items separated by a space)	

Figure 24 – Create NPA_1613 for outgoing calls to national numbers

Create Route List Block

Create RLI_10 for outgoing calls (Use route_100 and DMI_10), figure 25

NORTEL CS 1000 ELEMENT MANAGER Help

Data Entry of a Route List Block

Route List Block Index: 10

Input Description	Input Value
Entry Number for the Route List (ENTR):	0
Local Termination entry (LTER):	<input type="checkbox"/>
Route Number (ROUT):	100
Skip Conventional Signaling (SCNV):	<input type="checkbox"/>
Use Tone Detector (TDET):	<input type="checkbox"/>
Time of Day Schedule (TOD):	0
Entry is a VNS Route (VNS):	<input type="checkbox"/>
Conversion to LDN (CNV):	<input type="checkbox"/>
Expensive Route (EXP):	<input type="checkbox"/>
Facility Restriction Level (FRL):	0 (0 - 7)
Digit Manipulation Index (DMI):	10
ISL D-Channel Down Digit Manipulation Index (ISDM):	0 (0 - 999)
Free Calling Area Screening Index (FCI):	0

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Figure 25- RLB for Outgoing calls

Create Local Steering Code

Create LSC_585 to terminate the incoming calls (Use DMI_3); Figure 26

NORTEL CS 1000 ELEMENT MANAGER Help | Logout

Local Steering Code

Input Description	Input Value
Local Steering Code (LSC):	585
Digit Manipulation Index for LSC (DMI):	3
Number of digits to be deleted (DEL):	(1 - 7)

Submit Refresh Delete Cancel

Figure 26 – Create LCS_585 for incoming call

Create Digit Manipulation Block

DMI_10: Digit Manipulation Block configuration for Outgoing calls; figure 27

The screenshot shows the Nortel CS 1000 Element Manager interface. The left sidebar contains a tree view with categories like Peripheral Equipment, IP Network, Interfaces, Engineered Values, Emergency Services, Geographic Redundancy, Software, Customers, Routes and Trunks, Dialing and Numbering Plans, Phones, and Tools. The main content area is titled 'Digit Manipulation Block'. It displays a table with two columns: 'Input Description' and 'Input Value'. The table contains the following rows:

Input Description	Input Value
Digit Manipulation Index numbers (DMI):	10
Number of leading digits to be Deleted (DEL):	0 (0 - 19)
Insert (INST):	
IP Special Number (ISPN):	<input type="checkbox"/>
Call Type to be used by the manipulated digits (CTYP):	NPA (NPA)

Below the table are four buttons: Submit, Refresh, Delete, and Cancel. At the bottom, there is a copyright notice: Copyright © 2002-2009 Nortel Networks. All rights reserved.

Figure 27 – Digit Manipulation for Outgoing calls

DMI_3: Digit Manipulation Block configuration for Incoming calls; figure 28

The screenshot shows the Nortel CS 1000 Element Manager interface, similar to Figure 27, but for Incoming calls. The left sidebar is the same. The main content area is titled 'Digit Manipulation Block'. It displays a table with two columns: 'Input Description' and 'Input Value'. The table contains the following rows:

Input Description	Input Value
Digit Manipulation Index numbers (DMI):	3
Number of leading digits to be Deleted (DEL):	6 (0 - 19)
Insert (INST):	
IP Special Number (ISPN):	<input type="checkbox"/>
Call Type to be used by the manipulated digits (CTYP):	NPA (NPA)

Below the table are four buttons: Submit, Refresh, Delete, and Cancel. At the bottom, there is a copyright notice: Copyright © 2002-2009 Nortel Networks. All rights reserved.

Figure 28 – Digit Manipulation for Incoming calls

Configure on CS1000 Voicemail System (Call Pilot)

Configuration Details on CallPilot Manager

Configure CS1000E switch on Call Pilot configuration by entering:

- CS1000 Call Server IP address
- Create Multimedia Chanel for communication between CS1000 and Callpilot system

The screenshot displays the Nortel CallPilot Manager web interface. The top navigation bar includes links for Home, User, System, Maintenance, Messaging, Tools, and Help. The current page is titled "Configuration Wizard: M1 Switch Information" and includes buttons for Back, Next, Cancel, and Help. The main content area is titled "Meridian 1 Switch Information:" and provides instructions on updating channel settings. It shows the "STI Board 1 (201i in slot 01)" configuration. The "Switch Type" is set to "M1". The "Switch Customer Number" is 0. The "Switch IP Address" is 192.168.10.5. The "Symposium Call Center Server CLAN IP Address" is 192.168.10.50. A table lists the channels for the STI Board 201i, showing channel names, TN, Key0, Key1, Channel Allocation, and Class ID. The table includes channels for IVR and Multimedia. The bottom of the interface shows a sidebar with links for Call Server Initialization, Date and Time, Logs and reports, Security, Passwords, Policies, and Login Options. The footer includes the copyright notice "Copyright © 2002-2007 Nortel Networks. All rights reserved."

Location: Configuration Wizard → M1 Switch Information

Configuration Wizard: M1 Switch Information

Back Next Cancel Help

Meridian 1 Switch Information:

Channel information for each Link is displayed below. Click on a link to update its channel settings.

[STI Board 1 \(201i in slot 01\)](#)

[Link STI01-001](#)

[Link STI01-002](#)

Switch Type: ☒ M1 ☐ M1 Option 11

Switch Customer Number: 0

Switch IP Address: 192 . 168 . 10 . 5

☐ Enable Symposium Call Center Server Integration

Symposium Call Center Server CLAN IP Address: 192 . 168 . 10 . 50

STI Board 201i Board ID 68157440

Link STI01-001

#	Channel Name	TN	Key0	Key1	Channel Allocation	Class ID
1	STI01-001-001	0.0.9.0	3415	3301	IVR	
2	STI01-001-002	0.0.9.1	3416	3406	IVR	
3	STI01-001-003	0.0.9.2	3203	3303	Multimedia	
4	STI01-001-004	0.0.9.3	3204	3304	Multimedia	

Local DDD number to be recognized (LDDD):

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Figure 29 – CS1000 switch configuration on CallPilot Manager

Go to Maintenance pull down menu, select **Channel Monitor** to check status of the newly created multimedia channels on Call Pilot to see if the communication between Callpilot and CS1000 has been established, Figure 30.

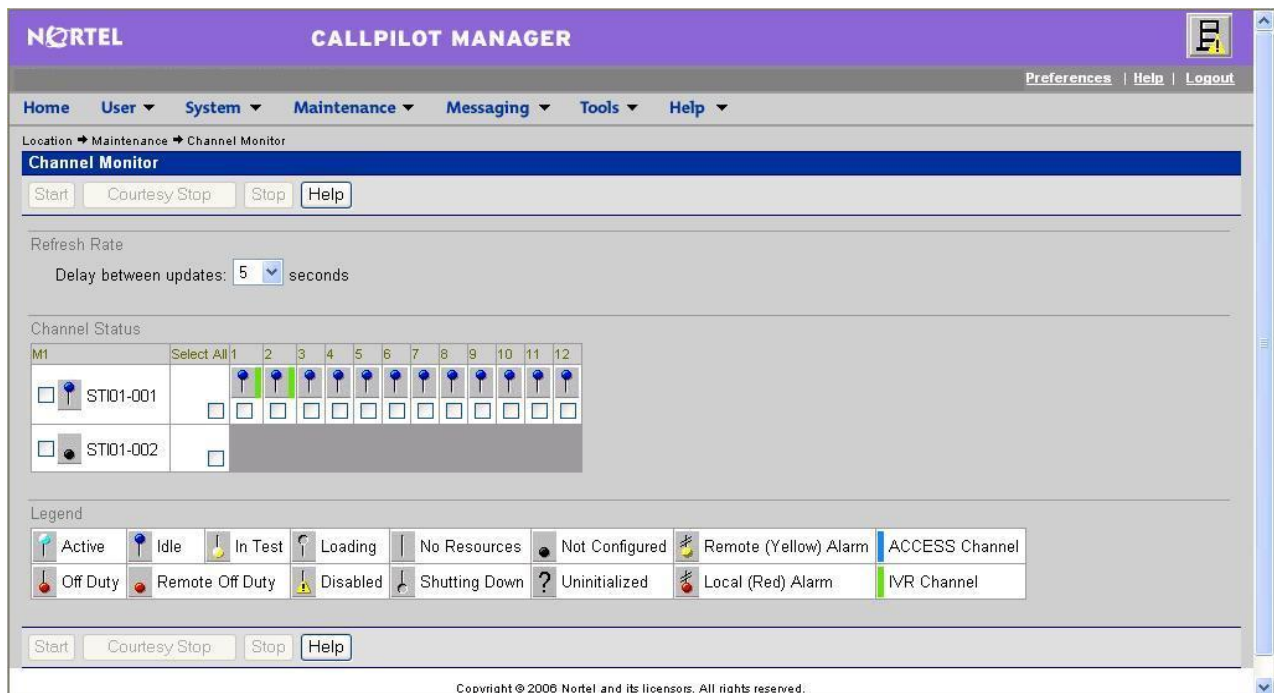


Figure 30 – Channel Monitor

Create Service DN for Voice Messaging system, Figure 31

NORTEL CALLPILOT MANAGER						
Home User System Maintenance Messaging Tools Help						
Location → System → Service Directory Number						
Service Directory Number						
Service Directory Number						
New Delete Selected Help						
#	Service DN	App Name	Media Type	Min Channels	Max Channels	Comments
1	3111	Voice Messaging	Voice	0	Default Max.	
2	3222	Express Voice Messaging	Voice	0	Default Max.	
5	OUTBOUND10	AMIS Networking	Voice	0	Default Max.	
6	OUTBOUND11	Remote Notification	Voice	0	Default Max.	
7	OUTBOUND15	Multi-delivery to Fax	Fax	0	Default Max.	
8	OUTBOUND18	Desktop Telephony Agent	Voice	0	Default Max.	
9	OUTBOUND23	SCCS VPE	Voice	0	Default Max.	
10	OUTBOUND25	Conferencing Outcalling	Voice	0	Default Max.	
11	OUTBOUND55	Enterprise Diagnostics	Voice	0	Default Max.	
12	OUTBOUND6	Admin Agent	Voice	0	Default Max.	
13	OUTBOUND7	Delivery To Telephone	Voice	0	Default Max.	
14	OUTBOUND8	Delivery To Fax	Fax	0	Default Max.	
15	OUTBOUND88	SCCS IVR	Voice	0	Default Max.	
16	OUTBOUND9	Enterprise Networking	Voice	0	Default Max.	

Figure 311 – Service Directory Number Page

Voicemail System (CallPilot) configuration detail on CS1000E Call Server

Configure CS1000E for voicemail system Call Pilot

Configure Voice messaging service DN 3111 on CS1000E

>ld 23

ACD DNS

REQ prt

TYPE CDN

CUST 0

CDN 3111

FRRT

SRRT

FROA NO

UUI NO

MURT

CDSQ NO

DFDN 3109

NAME NO

CMB NO

CEIL 2047

OVFL NO

TDNS NO

AACQ NO

CNTL NO

VSID

HSID

Configure ACD Agent #1 3110:

>ld 23

ACD DNS

REQ prt

TYPE ACD

CUST 0

ACDN 3110

MWC YES

MAXP 12

SDNB NO

BSCW NO

ISAP NO

AACQ YES

ASID 16

SFNB

USFB 1 3 4 5 6

CALB 1 3 4 5 6 8 11

RGAI NO

ACAA NO
FRRT
SRRT
NRRT
FROA NO
CALP POS
ICDD NO
NCFW
FNCF NO
CWTT NONE
HMSB YES
ACPQ NO
FORC NO
RTQT 0
SPCP NO
OBTN NO
RAO NO
CWTH 1
NCWL NO
BYTH 0
OVTH 2047
TOFT NONE
HPQ NO
OCN NO
OVDN
IFDN
OVBV LNK LNK LNK LNK
EMRT
MURT
RTPC NO
NRAC NO
RAGT 4
DURT 30
RSND 4
FCTH 20
CRQS 100
CCBA NO
IVR YES
TRDN NONE
ABR NO
OBSC NO
OBPT 5
CWNT NONE
Configure ACD Agent#2
>ld 23

REQ prt
TYPE acd
CUST 0
ACDN 3109
TYPE ACD
CUST 0
ACDN 3109
MWC NO
DSAC NO
MAXP 1
SDNB NO
BSCW NO
ISAP NO
AACQ NO
RGAI NO
ACAA NO
FRRT
SRRT
NRRT
FROA NO
CALP POS
ICDD NO
NCFW
FNCF NO
CWTT NONE
HMSB YES
ACPQ NO
FORC NO
RTQT 0
SPCP NO
OBTN NO
RAO NO
CWTH 1
NCWL NO
BYTH 0
OVTH 2047
TOFT NONE
HPQ NO
OCN NO
OVDN
IFDN
OVBU LNK LNK LNK LNK
EMRT
MURT
RTPC NO

NRAC NO
RAGT 4
DURT 30
RSND 4
FCTH 20
CRQS 100
CCBA NO
IVR NO
ABR NO
OBSC NO
OBPT 5
CWNT NONE

Output configuration details from CS1000 and Callpilot

Using the command line interface , output of some of configured Customer Data Block and configuration record details, which have been created in section 4.2 is shown below for your reference

Overlay 15 - Customer Data Block

REQ: PRT
TYPE CDB
CUST 00
AML_DATA
OPT DNX
VSID
GP02
GP03
GP04
GP05
GP06
GP07
GP08
GP09
GP10
GP11
GP12
GP13
GP14
GP15
ANI_DATA
ANAT 4227
ANLD 123
M911_PANI NO
ATT_DATA

OPT ABDD AHD BIND BIXA BLA BOHD DNCA DRE
DNX DRE FACD IC1 XTG XDP XLF XBL
FKA MCTD NCD CUI MWUD LOD PSD RECA
REA SYD SLD SIAD THPD ATDA

ATDN 7
NCOS 0
CWUP NO
CWCL 0 0
CWTM 0 0
CWBZ NO NO
EFLL 0
MATT NO
RTIM 30 30 30
ATIM 0
AQTT 30
AODN
SPVC 00
SBLF NO
RTSA RSAD
SACP NO
ABDN NO
IRFR NO
XRFR NO
ADHT 0
AFNT 0
AFBT 0
IDBZ NO
PBUZ 02 10
ICI 00
ICI 01
ICI 02
ICI 03
ICI 04
ICI 05
ICI 06
ICI 07
ICI 08
ICI 09
RICI
PAGE 002
AWU_DATA
AWU NO
CAS_DATA
CAS NO
CCS_DATA

CCRS UNR
 ECC1 UNR
 ECC2 UNR
 CNCS 0
 PELK NO
 CDR_DATA
 CDR YES
 IMPH NO
 OMPH YES
 AXID YES
 TRCR NO
 CDPR NO
 ECDR NO
 BDI YES
 OTCR NO
 PORT
 CNI DGTS
 BCAP NO
 CHLN 1
 FCAF NO
 FCR_DATA
 NFCR YES
 MAXT 100
 OCB1 255
 OCB2 255
 OCB3 255
 IDCA YES
 DCMX 100
 FFC_DATA
 CCRS UNR
 SCPL 0
 FFCS NO
 STRL 0
 STRG
 ADLD 0
 MFAC *
 FTR_DATA
 DAPCPREFIX TABLE NO: 00 **
 UNKN**INTL**NATL**ESPN**LOCL**ELOC**ECDP**
 UNKN*
 E164* 00 0
 PRIV*
 E163* 00 0
 TELX*
 X121*

NATL*

OPT ABDD AHD BIND BIXA BLA BOHD CFO CFRD
COX CPA CTD DBD DNCA DNX DSX DRE
DSTD FACD HTU HVD XBL IC1 XDP XLF
IHD XTG FKA LOD LRA MCI MCTD CUI
MWUD NCD PCMD PSD PVCA RECA REA RND
RTR RTD ROX SBD SDDE SIAD SLD SYD
THPD TTAD VOBD CCB D CWRD HLPD HRLD
CXOD BWTD

DGRP 0
IRNG NO
PKND 1
DNDL NO
SPRE
PREO 0
BPSS NO
SRCD 0000
EEST NO
EESD NO
TTBL 0
MUS YES

PAGE 003

MUSR 50
HCC NO
ALDN
RECD NO
PORT 0
STCB NO
NSCP NO
TFDR NO
RPA NO
MCDC NO
NAUT NO
IDEF NO
MTAR NO
LEND NO
MSCD NO
CPCI NO
ARDL_ATTEMPT 30
CONF_DSP
CNFFIELD NO
CNF_NAME CONF
INTFIELD NO

INT_NAME I
EXTFIELD NO
EXT_NAME E
BSFE NO
ASPCT 000
FXS NO
DFLT_LANG ENG
STS_MSG
MSG01 Please leave message
MSG02 Back to work
MSG03 In a meeting
MSG04 On a conference call
MSG05 At lunch
MSG06 Busy call
MSG07 Out of the office today
MSG08 On a business trip
MSG09 Project deadline today
MSG10 Will reply after
VO_ALO NO
PCA ON
TPDN
BFS_CFW YES
VO_CUR_ZONE_ZDM NO
VO_CUR_ZONE_TD NO
ICP_DATA
ICP NO
IMS_DATA
IMS NO
INT_DATA
ACCD OVF OVF OVF ATN
CTVN OVF OVF OVF ATN
MBNR OVF OVF OVF ATN
CTRC OVF NAP OVF NAP
CLDN NAP OVF NAP NAP
NINV OVF OVF OVF ATN
NITR OVF OVF OVF ATN
NRES OVF OVF OVF ATN
NBLK OVF OVF OVF ATN
MFVOOVF OVF OVF ATN
MFVN OVF OVF OVF ATN
MFCG OVF OVF OVF ATN
PAGE 004
LCKT BSY BSY BSY BSY
RCLE ATN OVF ATN ATN
CONG OVF

DLT OVF
LLT OVF
DNDT BSY
ESAM OVF
LDN_DATA
OPT XLDN
DLDN YES
LDN0 2000
LDA0
LDN1
LDA1
LDN2
LDA2
LDN3
LDA3
LDN4
LDA4
LDN5
LDA5
LDBZ
ICI 00
ICI 01
ICI 02
ICI 03
ICI 04
ICI 05
ICI 06
ICI 07
ICI 08
ICI 09
MON_DATA
USBM NO
MPO_DATA
FMOP
RGNA STD STD
AOCS DIS ATN
RCY1 06
RCY2 04
RALL NO
CDTO 14
IFLS NO
MHLD NO
PCDS
CNFD 1
TGLD 2

DISD 3
CCDO NO
AFCO NO
ACNS NO
NET_DATA
OPT RTD
AC1 NPA SPN LOC
AC2 INTL NXX
FNP YES
ISDN YES
VPNI 1
PNI 1
PINX_DN
MBG 0

PAGE 005
BSGC 65535
PFX1
PFX2
HLOC 521
LSC
RCNT 5
PSTN NO
TNDM 15
PCMC 15
SATD 1
OCLI NO
TIDM NO
DASC
ROPT NRO
DITI YES
TRNX NO
EXTT NO
FTOP FRES
APAD 0 0
VNR NO
NIT 8
NAS_ATCL YES
NAS_ACTV NO
FOPT 6
CNDN
CNAT
PCAT
CNIP YES
DMWM NO

MWNS NO
CNTC
NATC
INTC
NIT_DATA
NIT1
TIM1
NIT2
TIM2
NIT3
TIM3
NIT4
TIM4
RPNS NO
ENS NO
OAS_DATA
ODN0
ODN1
ODN2
ODN3
ODN4
ODN5
ODN6
ODN7
ODN8
ODN9
ASTM 30
HDOPT 0
HDTM 30
RDR_DATA
OPT CFO CFRD DSTD PVCA CWRD MCI
FNAD HNT
FNAT HNT
PAGE 006
FNAL HNT
CFTA NO
CCFWDN
CFN0 3
CFN1 3
CFN2 3
DFN0 3
DFN1 3
DFN2 3
DNDH NO
MDID NO

NDID NO
MWFB NO
TRCL 0
DFNR 0
CRT0 00 00 00 00
CRT1 00 00 00 00
CRT2 00 00 00 00
CRT3 00 00 00 00
DAY0
DAY1
DAY2
DAY3
HOLIDAY0
HOLIDAY1
HOLIDAY2
HOLIDAY3
ROA_DATA
OPT ROX
RICI
TIM_DATA
FLSH 45 896
PHDT 30
DIND 30 32 30
DIDT 14 16 14
LDTT 6
DLAT 0
BOTO 14
DBRC 60
RTIM 30 30 30
ATIM 0
AQTT 30
ADLD 0
AFNT 0
NFNA 0
ADHT 0
HWTT 300
NIT 8
FOPT 6
ARDL_ACCEPT 20
ARDL_RETRY 30
TST_DATA

Overlay 17 – Configuration Record

REQ PRT
TYPE CFN
ADAN HIST
SIZE 25000
USER MTC BUG
ADAN TTY 0
CTYP PTY
DNUM 0
PORT 0
DES PTY0
FLOW NO
USER MTC TRF SCH BUG OSN
XSM NO
TTYLOG 0
BANR NO
ADAN TTY 1
CTYP PTY
DNUM 1
PORT 1
DES PTY1
FLOW NO
USER MTC TRF SCH BUG OSN
XSM NO
TTYLOG 0
BANR NO
ADAN TTY 2
CTYP PTY
DNUM 2
PORT 2
DES PTY2
FLOW NO
USER MTC TRF SCH BUG OSN
XSM NO
TTYLOG 0
BANR NO
ADAN TTY 3
CTYP PTY
DNUM 3
PORT 3
DES PTY3
FLOW NO
USER MTC TRF SCH BUG OSN
XSM NO
TTYLOG 0
BANR NO

ADAN TTY 4
CTYP CPSI
DNUM 4
PORT 0
DES
BPS 9600
BITL 8
STOP 1
PARY NONE
FLOW NO
USER MTC TRF SCH BUG OSN
XSM NO
TTYLOG 0
BANR NO

ADAN TTY 5
CTYP CPSI
DNUM 5
PAGE 001
PORT 1
DES
BPS 9600
BITL 8
STOP 1
PARY NONE
FLOW NO
USER MTC TRF SCH BUG OSN
XSM NO
TTYLOG 0
BANR YES

ADAN ELAN 16 (Configuration for CallPilot)

CTYP ELAN
DES CPilot
N1 512
ADAN DCH 100
CTYP DCIP
DES VoIP
USR ISLD
ISLM 4000
SSRC 1800
OTBF 32
NASA YES
IFC SL1
CNEG 1
RLS ID 5

RCAP ND2 MWI (Configuration for CallPilot)

MBGA NO
H323
OVLN NO
OVLS NO
ADAN DCH 101
CTYP DCIP
DES Enterprise
USR ISLD
ISLM 4000
SSRC 1800
OTBF 32
NASA NO
IFC SL1
CNEG 1
RLS ID 25
RCAP ND2 MWI
MBGA NO
H323
PAGE 002
OVLN NO
OVLS NO
PARM
LPIB 3500
HPIB 3500
500B 2000
SL1B 255
DTIB 35
DFOB 4
NCR 20000
MGCR 25
CSQI 255
CSQO 255
TUBO NO
NCPU 2
CFWS NO
PCML A
ALRM YES
ERRM ERR BUG AUD
DTRB 100
ABCD NO
TMRK 128
FCDR OLD
PCDR NO
TPO NO
TSO NO

CLID NO
DUR5 NO
MLDN NO
MARP YES
IPIE NO
FRPT NEFR
DCUS NULL
DTDT NO
MSCL 0
PMSI
MANU PMS1
PMCR 0
PORT NONE
NDIS 20
OCAC NO
MTRO MR
SBA_ADM_INS 000
SBA_USER 512
BCAP SPEECH
IDLE_SET_DISPLAY
ICON NO
MSEC ON
MSSD MSBT
NKEY 31
TKEY 24
CEQU
MPED 8D
TERM
REMO
TERD
REMD
TERQ
REMQ
SUPL V000 V096 V100 V200
SUPC
PAGE 003
SUPF
DDCS MG_CARD
DTCS
XCT
CONF
MGTDs IPMG IPMG_TYPE
126 000 0 MGC
MGCONF IPMG PORTS IPMG_TYPE
127 000 0 30 MGC

MFSD * 126
APVL
MISP MG_CARD
SYNM 0
EXT0 3PE
EXT1 3PE
MCFN 011 MB
OVLY
SID 0
BKGD 044
PBXH X
TODR 00
DROL 030 032 045 135 137
MID_SCPU NO
CY45 00
MULTI_USER OFF
VAS
VSID 016
DLOP
ELAN 016
SECU NO
INTL 0001
MCNT 9999
VSID 022
DLOP
ELAN 022
SECU YES
INTL 0001
MCNT 9999
VSID 034
DLOP
ELAN 034
SECU YES
INTL 0001
MCNT 9999

VSID 035
DLOP
ELAN 035
SECU NO
INTL 0001
MCNT 9999

VSID 038
DLOP

ELAN 038
SECU YES
INTL 0001
MCNT 9999
PAGE 004
ATRN
CODE 0
SOLR 12
ROLR +45.00
AOLR +45.00
TOLR -45.00
AGCD NO
VOLR NO
HRLR +42.00
HTLR -44.00
ESA
LIS EXT/DM
DYNAMIC_ELIN_TIMEOUT 180
DYNAMIC_ELIN_REUSE YES
EXT_DM_UPDT_TIMEOUT 15

CS1K Tandem Configuration

This configuration is for the deployment model of 2 or more CS1000 with Frontier communication system. Represent here is for 2 CS1000s configuration.

Configure CS1000E A

1. Create IP Telephony Node on CS1000E

This section describes the steps for creating Node ID (1001) in CS 1000 network.

Launch Element Manager through the IE browser (in IE address bar, type IP address of the Node IP or TLAN of Signaling Server).

- Input Node ID and Click on **Add** button.
- Enter TLAN, ELAN IP address of Signaling Server.

Node 1001 was added to be configured as the SIP gateway to the Enterprise services.

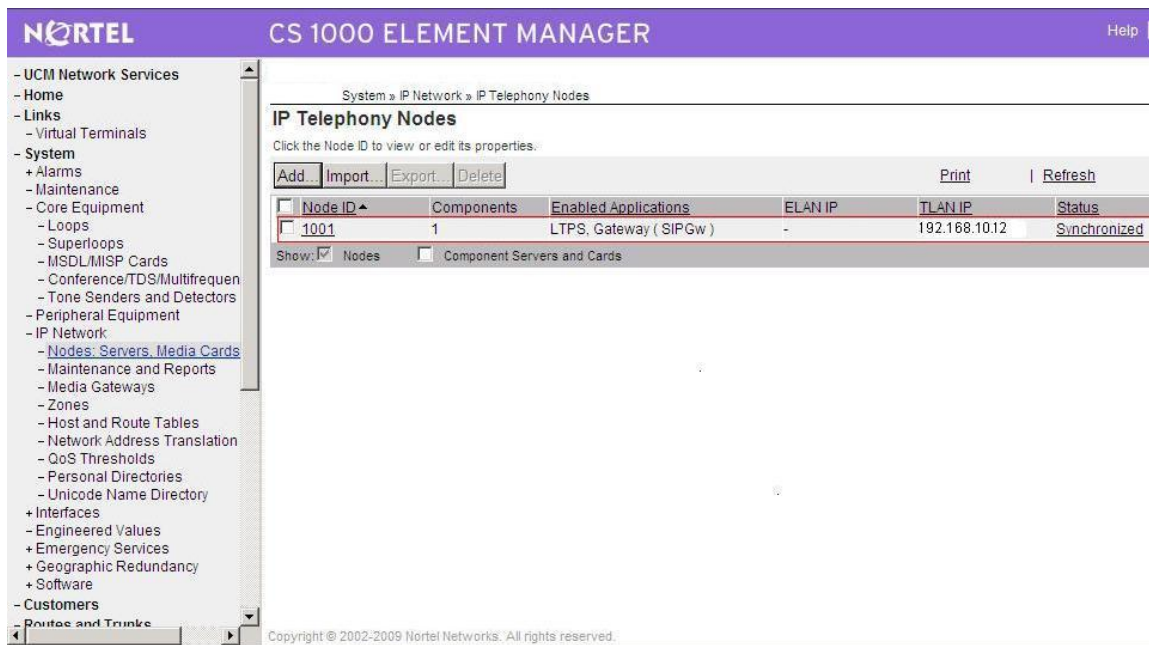


Figure 32 – Creating a node on CS1000 A

The node IP information is added. For the primary proxy, Enter the IP address of the SIP Proxy Server (SPS). Use UDP port 5060 for SIP communication, figure 33

NORTEL CS 1000 ELEMENT MANAGER Help

- UCM Network Services
- Home
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 - + Maintenance
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 - + Peripheral Equipment
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 - Nodes, Servers, Media Cards
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 - Routes and Trunks
 - D-Channels
 - Digital Trunk Interface
- Dialing and Numbering Plans
 - Electronic Switched Network
 - Flexible Code Restriction
 - Incoming Digit Translation
- Phones

System » IP Network » IP Telephony Nodes » Node Details

Node Details (ID: 1001 - LTPS, PD, Gateway (SIPGw))

Node ID: * (0-9999)

Call Server IP Address:

Telephony LAN (TLAN)

Node IP Address:

Subnet Mask:

Embedded LAN (ELAN)

Gateway IP address:

Subnet Mask:

IP Telephony Node PropertiesApplications (click to edit configuration)

- [Voice Gateway \(VGW\) and Codecs](#)
- [Quality of Service \(QoS\)](#)
- [Terminal Proxy Server \(TPS\)](#)

* Required Value. Save Can

Associated Signaling Servers & Cards

Add Remove Make Leader

[Print](#) | [Refresh](#)

<input type="checkbox"/> Hostname	Type	Deployed Applications	ELAN IP	TLAN IP	Role
<input type="checkbox"/> nd1-car1	Signaling Server	LTPS, Gateway, PD	192.168.100.151	192.168.10.147	Leader

Note: Only server(s) that are not part of any other IP telephony node and deployed application(s) that match the service(s) selected for this node are available in the servers list.

Figure 33 – CS1000 Node Detail Settings

In the signaling server properties, the Line TPS will be enabled if this signaling server will be used for IP set registration. If the role of this server is SIP gateway only, then this can be left unchecked as shown in figure 34.

NOTEL CS 1000 ELEMENT MANAGER Help |

System » IP Network » IP Telephony Nodes » Node Details » Virtual Trunk Gateway Configuration

Node ID: 1001 - Virtual Trunk Gateway Configuration Details

General	SIP Gateway Settings	SIP Gateway Services
Vtrk Gateway Application: <input checked="" type="checkbox"/> Enable gateway service on this Node		
General Vtrk Gateway Application: <input type="text" value="SIP Gateway (SIPGw)"/> SIP Domain name: <input type="text" value="interop.com"/> Local SIP Port: <input type="text" value="5060"/> *(1 - 65535) Gateway endpoint name: <input type="text" value="car1_ss2"/> Gateway password: <input type="password"/> Enable failsafe NRS: <input type="checkbox"/>		Virtual Trunk Network Health Monitor <input checked="" type="checkbox"/> Monitor IP Addresses (listed below) Information will be captured for the IP addresses listed below. Monitor IP: <input type="text"/> <input type="button" value="Add"/> Monitor addresses: <div style="border: 1px solid black; height: 40px; width: 100%;"></div> <input type="button" value="Remove"/>
SIP Gateway Settings TLS Security: <input type="text" value="Security Disabled"/> Port: <input type="text" value="5061"/> *(1 - 65535) Number of Byte Re-negotiation: <input type="text" value="0"/> Options: <input type="checkbox"/> Client Authentication <input checked="" type="checkbox"/> X509 certificate authority		
Proxy Or Redirect Server: Primary TLAN IP Address: <input type="text" value="192.168.10.60"/> Port: <input type="text" value="5060"/> *(1 - 65535) Transport protocol: <input type="text" value="UDP"/> Options: <input checked="" type="checkbox"/> Support registration <input type="checkbox"/> Primary CDS Proxy		Secondary TLAN IP Address: <input type="text" value="0.0.0.0"/> Port: <input type="text" value="5060"/> *(1 - 65535) Transport protocol: <input type="text" value="UDP"/> Options: <input type="checkbox"/> Support registration <input type="checkbox"/> Secondary CDS Proxy
<p>* Required Value.</p> <p>Note: Changes made on this page will NOT be transmitted until the Node is also saved.</p> <div style="text-align: right;"> <input type="button" value="Save"/> <input type="button" value="Cancel"/> </div>		

Figure 34 – SIP Gateway Settings

2. Create D-channel (DCH)

- Launch Element Manager of CS 1000 6.0
- Choose D-Channels, enter D-channel number (i.e.: 101), select DCH for type
- Click Add to create DCH 101 as shown in figure 35; Also click on Options and edit the Remote Capabilities (RCAP). Enable MWI if CS1K hosted voice mail will be used.

CS 1000 ELEMENT MANAGER

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- Routes and Trunks
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- Dialing and Numbering Plans
- Electronic Switched Network
- Network Routing Service
- Flexible Code Restriction
- Incoming Digit Translation
- Tools
- Backup and Restore
- Call Server Initialization
- Date and Time
- Logs and reports
- Security
- Passwords
- Policies
- Login Options

Managing: 192.168.10.5
Routes and Trunks » D-Channels » D-Channels 101 Property Configuration

D-Channels 101 Property Configuration

- Basic Configuration

Input Description	Input Value
Action Device And Number (ADAN) (TYPE)	DCH
D channel Card Type (CTYP)	DCIP
Designator (DES)	VoIP
Recovery to Primary (RCVP)	<input type="checkbox"/>
PRI loop number for Backup D-channel (BCHL)	
User (USR)	Integrated Services Signaling Link Dedicated (ISLD)
Interface type for D-channel (IFC)	Meridian Meridian1 (SL1)
Country (CNTY)	ETS 300 =102 basic protocol (ETSI)
D-Channel PRI loop number (DCHL)	
Primary Rate Interface (PRI)	<input type="text"/> <button>more PRI</button>
Secondary PRI2 loops (PRI2)	<input type="text"/>
Meridian 1 node type (SIDE)	Slave to the controller (USR)
Release ID of the switch at the far end (RLS)	25
Central Office switch type (CO_TYPE)	100% compatible with Bellcore standard (STD)
Integrated Services Signaling Link Maximum (ISLM)	4000 Range: 1 - 4000
Signaling Server Resource Capacity (SSRC)	1800 Range: 0 - 4000

+ Basic options (BSCOPT)
+ Advanced options (ADVOPT)

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Figure 35 – D-Channels Property Configuration

3. Create Route

Create route 101 using DCH 101 for SIP trunks as shown in Figure 36

Nortel CS 1000 ELEMENT MANAGER

Managing: **192.168.10.5**
Routes and Trunks » Routes and Trunks » Customer 0, Route 101 Property Configuration

Customer 0, Route 101 Property Configuration

- Basic Configuration

Input Description	Input Value
Route Data Block (RDB) (TYPE)	RDB
Customer number (CUST)	00
Route Number (ROUT)	101
Designator field for trunk (DES)	ENTERPRISE
Trunk Type (TKTP)	TIE
Incoming and Outgoing trunk (ICOG)	Incoming and Outgoing (IAO)
Access Code for the trunk route (ACOD)	8101
Trunk type M911P (M911P)	<input type="checkbox"/>
The route is for a virtual trunk route (VTRK)	<input checked="" type="checkbox"/>
Zone for codec selection and bandwidth management (ZONE)	255 <small>Range: 0 - 255</small>
Node ID of signaling server of this route (NODE)	1001 <small>Range: 0 - 9999</small>
Protocol ID for the route (PCID)	SIP (SIP)
Print Correlation ID in CDR for the route (CRID)	<input type="checkbox"/>
Integrated Services Digital Network option (ISDN)	<input checked="" type="checkbox"/>
Mode of operation (MODE)	Route uses ISDN Signaling Link (ISLD)
D channel number (DCH)	101 <small>Range: 0 - 254</small>
Interface type for route (IFC)	Meridian M1 (SL1)

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Figure 36 – Route Property Configuration

Configure Route 101 for SIP trunks continue as shown in figure 40

Nortel CS 1000 ELEMENT MANAGER

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- Home

- Links

- Virtual Terminals
- Bookmarks

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 - Media Gateways
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 - Host and Route Tables
 - Network Address Translation
 - QoS Thresholds
 - Personal Directories
- + Interfaces
- + Engineered Values
- + Emergency Services
- + Software

- Customers

- Routes and Trunks

- Routes and Trunks
- D-Channels
- Digital Trunk Interface

- Dialing and Numbering Plans

- Electronic Switched Network
- Network Routing Service
- Flexible Code Restriction
- Incoming Digit Translation

- Tools

- + Backup and Restore
- Call Server Initialization
- Date and Time
- Logs and reports

- Security

- Records

- D channel number (DCH) 101 Range: 0 - 254

- Interface type for route (IFC) Meridian M1 (SL1)

- Private Network Identifier (PNI) 00001 Range: 0 - 32700

- Network Calling Name Allowed (NCNA) ☒

- Network Call Redirection (NCRD) ☐

- Recognition of DTI2 ABCD FALT signal for ISL (FALT) ☐

- Channel Type (CHTY) B-channel (BCH)

- Call Type for outgoing direct dialed TIE route (CTYP) Unknown Call type (UKWVN)

- Insert ESN Access Code (INAC) ☒

- Integrated Service Access Route (ISAR) ☐

- Display of Access Prefix on CLID (DAPC) ☐

- Mobile Extension Route (MBXR) ☐

- Basic Route Options

Input Description	Input Value
Billing Number Required (BILN)	<input type="checkbox"/>
Call Detail Recording (CDR)	<input type="checkbox"/>
Controls or timers (CNTL)	<input type="checkbox"/>
Conventional (Tie trunk only) (CNVT)	<input type="checkbox"/>
Incoming DID Digit Conversion on this route (IDC)	<input type="checkbox"/>
Multifrequency Compelled or MFC Signaling (MFC)	No MFC (NO)
Process Notification Networked Calls (PNNC)	<input type="checkbox"/>

+ Network Options

+ General Options

+ Advanced Configurations

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Figure 37 – Route Property Configuration Details (cont.)

4. Create Trunk (figure 38)

Since Media security is not supported under Frontier system, Disable Media Security (MSNV) at the Trunk level as show in figure 39

Nortel CS 1000 ELEMENT MANAGER

Help | Logout

Managing: 192.168.10.5

Routes and Trunks > Routes and Trunks > Customer 0, Route 101, Trunk 1 Property Configuration

Customer 0, Route 101, Trunk 1 Property Configuration

- Basic Configuration

Input Description	Input Value
Trunk data block (TYPE)	IFTI
Terminal Number (TN)	100 0 01 00
Designator field for trunk (DES)	CARRIER
Extended Trunk (XTRK)	VTRK
Route number, Member number (RTMB)	101 1
Level 3 Signaling (SIGL)	
Card Density (CDEN)	8D
Start arrangement Incoming (STR)	Immediate (IMM)
Start arrangement Outgoing (STRO)	Immediate (IMM)
Trunk Group Access Restriction (TGAR)	0
Channel ID for this trunk (CHID)	101
Increase or decrease the member numbers (INC)	Increase channel and member number (YES)
Class of Service (CLS)	Edit

+ Advanced Trunk Configurations

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Figure 38 – Trunk Configuration Details

NORTEL CS 1000 ELEMENT MANAGER Help | Logout

Managing: **192.168.10.6**
 Routes and Trunks > Routes and Trunks > Customer 0, Route 101, Trunk 1 Property Configuration > Class of Service Configuration

Class of Service Configuration

Input Description	Input Value
- ACD Priority (CLS)	ACD Priority not required (APN)
- Analog Semi-Permanent Connections (CLS)	Analog Semi-Permanent Connections Denied (SPCD)
- ARF Supervised COT (CLS)	
- Barring (CLS)	
- Battery Supervised COT (CLS)	
- Busy Tone Supervised COT (CLS)	
- Calling party (CLS)	Calling party Denied (CND)
- Central Office Ringback (CLS)	
- Centrex Switchhook Flash (CLS)	Centrex Switchhook Flash Denied (THFD)
- Dial Pulse (CLS)	Digitone (DTN)
- DTR PAD value (CLS)	
- Echo Canceling (CLS)	Echo Canceling Denied (ECD)
- Hong Kong DTI (CLS)	
- Loop Break Supervised COT (CLS)	
- Make-break ratio for dial pulse (CLS)	10 pulses per second (P10)
- Manual Incoming (CLS)	Manual Incoming Denied (MID)
- Media Security (CLS)	Media Security Never (MSNV)
- Network Hook Flash Over M911P (CLS)	
- Polarity (CLS)	

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Figure 39 – Class of Services configuration Details (cont.)

5. Configure Dialing Plan for CS1000E_A

Create Location Code:

Create LOC 521 for basic outgoing calls to CS1000E_B (Use RLI_5; DMI_0); Figure 40

NORTEL CS 1000 ELEMENT MANAGER Help

Dialing and Numbering Plans > Electronic Switched Network (ESN) > Customer 00 > Numbering Plan (NET) > Access Code 2 > Location Code
 List > Location Code

Location Code

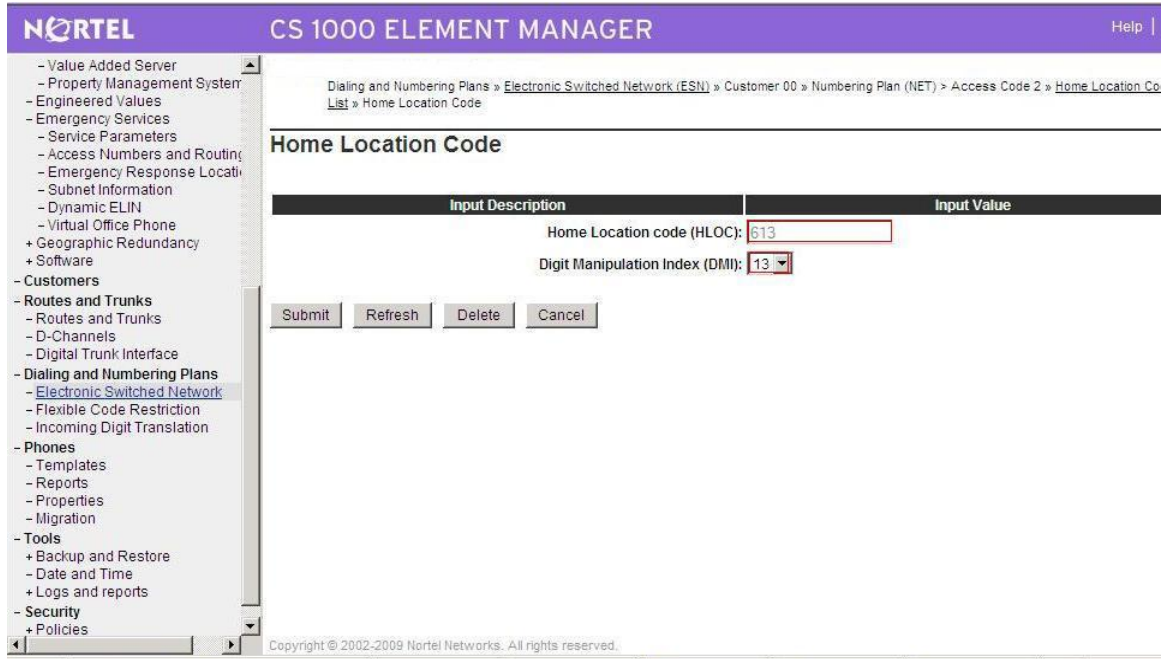
Input Description	Input Value
Location code (LOC):	521
Flexible Length (FLEN):	7 (0 - 10)
Route List Index (RLI):	5
Maximum 7 digit NPA code allowed (NPA):	
Maximum 7 digit NXX code allowed (NXX):	
Inhibit Time Out Handler (ITOH):	<input type="checkbox"/>
Incoming Trunk group Exclusion Index (ITE):	
Listed Directory Number (LDN):	123456789
Direct Inward Dial (DID):	<input type="checkbox"/>

Submit Refresh Delete Cancel

Figure 40 – LOC_521 for basic outgoing calls to CS1000EB

Create Home Location Code

Create HLOC_613 for incoming calls from CS1000E_B and outgoing calls to PSTN; Figure 41



NORTEL CS 1000 ELEMENT MANAGER Help

Dialing and Numbering Plans > Electronic Switched Network (ESN) > Customer 00 > Numbering Plan (NET) > Access Code 2 > Home Location Code

Home Location Code

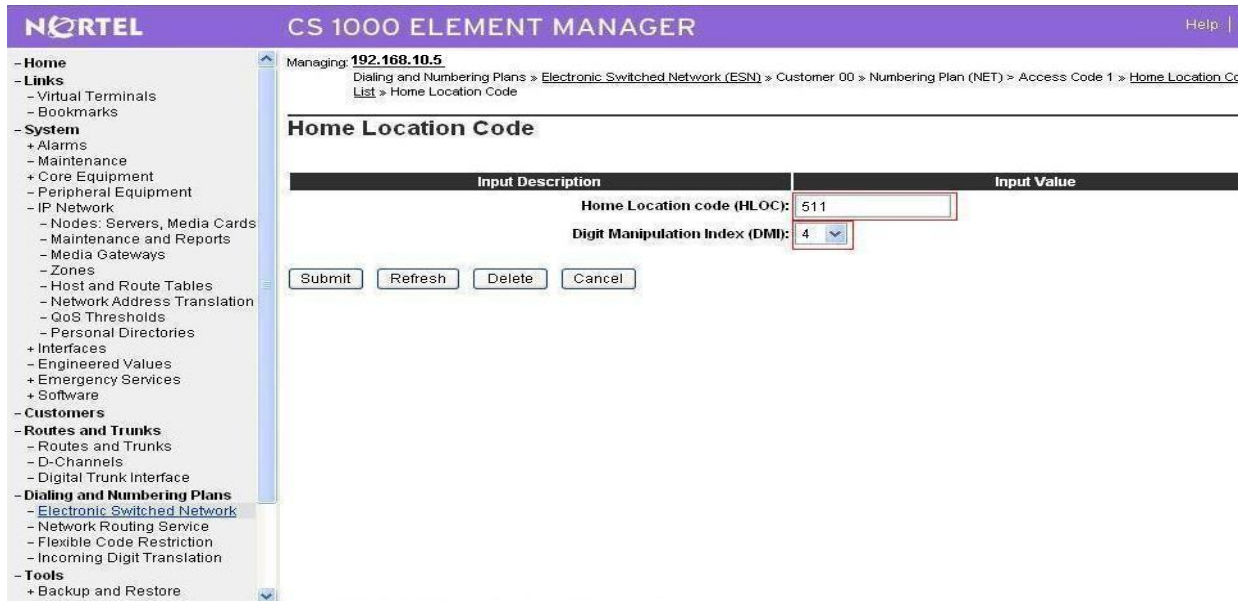
Input Description	Input Value
Home Location code (HLOC):	613
Digit Manipulation Index (DMI):	13

Submit Refresh Delete Cancel

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Figure 41 – HLOC: 613 to tandem calls from CS1000B to PSTN

Create HLOC_511 for basic incoming call from CS1000E_B (DMI_4); Figure 42



NORTEL CS 1000 ELEMENT MANAGER Help

Managing: 192.168.10.5

Dialing and Numbering Plans > Electronic Switched Network (ESN) > Customer 00 > Numbering Plan (NET) > Access Code 1 > Home Location Code

Home Location Code

Input Description	Input Value
Home Location code (HLOC):	511
Digit Manipulation Index (DMI):	4

Submit Refresh Delete Cancel

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Figure 42 – HLOC_511 to terminate calls from CS1000E_B

Create Distant Steering Code

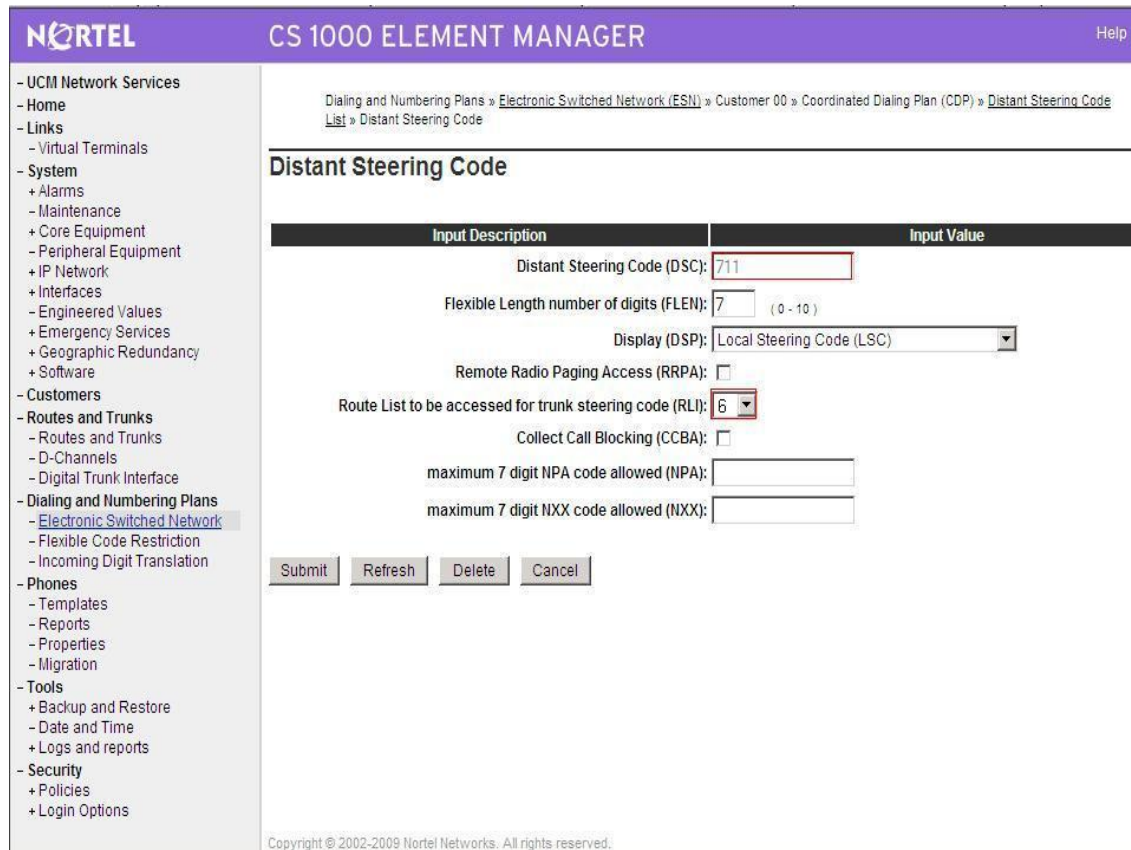
Create DSC_585 (RLI_6) to receive Calls from PSTN and tandem to CS1000E_B; Figure 43

Input Description	Input Value
Distant Steering Code (DSC):	585
Flexible Length number of digits (FLEN):	10 (0 - 10)
Display (DSP):	Local Steering Code (LSC)
Remote Radio Paging Access (RRPA):	<input type="checkbox"/>
Route List to be accessed for trunk steering code (RLI):	6
Collect Call Blocking (CCBA):	<input type="checkbox"/>
maximum 7 digit NPA code allowed (NPA):	
maximum 7 digit NXX code allowed (NXX):	

Submit Refresh Delete Cancel

Figure 43 – DSC_585 to receive calls from PSTN and tandem to CS1000E_B

Create DSC_711 to tandem calls from CS1000E_A to CS1000E_B; Figure 44



NORTEL CS 1000 ELEMENT MANAGER Help

Dialing and Numbering Plans » [Electronic Switched Network \(ESN\)](#) » Customer 00 » Coordinated Dialing Plan (CDP) » [Distant Steering Code List](#) » Distant Steering Code

Distant Steering Code

Input Description	Input Value
Distant Steering Code (DSC):	711
Flexible Length number of digits (FLEN):	7 (0 - 10)
Display (DSP):	Local Steering Code (LSC)
Remote Radio Paging Access (RRPA):	<input type="checkbox"/>
Route List to be accessed for trunk steering code (RLI):	6
Collect Call Blocking (CCBA):	<input type="checkbox"/>
maximum 7 digit NPA code allowed (NPA):	
maximum 7 digit NXX code allowed (NXX):	

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Figure 44 – DSC_711 to tandem calls to CS1000E_B.

Create Route List Block

Create RLI_5 for basic outgoing call to CS1000E_B (Use route 101, DMI_0); Figure 45

NORTEL CS 1000 ELEMENT MANAGER Help

Data Entry of a Route List Block

Route List Block Index: 5

Input Description	Input Value
Entry Number for the Route List (ENTR):	0
Local Termination entry (LTER):	<input type="checkbox"/>
Route Number (ROUT):	101
Skip Conventional Signaling (SCNV):	<input type="checkbox"/>
Use Tone Detector (TDET):	<input type="checkbox"/>
Time of Day Schedule (TOD):	0
Entry is a VNS Route (VNS):	<input type="checkbox"/>
Conversion to LDN (CNV):	<input type="checkbox"/>
Expensive Route (EXP):	<input type="checkbox"/>
Facility Restriction Level (FRL):	0 (0 - 7)
Digit Manipulation Index (DMI):	0
ISL D-Channel Down Digit Manipulation Index (ISDM):	0 (0 - 999)
Free Calling Area Screening Index (FCI):	0
Free Special Number Screening Index (FSNI):	0
Business Network Extension Route (BNE):	<input type="checkbox"/>
Strategy on Congestion (SBOC):	No Reroute (NRR)
- QSIG Alternate Routing Causes (COPT):	QSIG Alternate Routing Cause 1
ISDN Drop Back Busy (IDBB):	Drop Back Disabled (DBD)

Figure 45 – Create RLI_5 for simple outgoing calls to CS1000E_B

Create RLI_6 to for incoming calls from PSTN and outgoing calls to CS1000E_B (Use route 101, DMI_6); Figure 46

NORTEL CS 1000 ELEMENT MANAGER Help |

Dialing and Numbering Plans » Electronic Switched Network (ESN) » Customer 00 » Network Control & Services » Route List Blocks » Route List Block » Data Entry of a Route List Block

Data Entry of a Route List Block

Route List Block Index: 6

Input Description	Input Value
Entry Number for the Route List (ENTR):	0
Local Termination entry (LTER):	<input type="checkbox"/>
Route Number (ROUT):	101
Skip Conventional Signaling (SCNV):	<input type="checkbox"/>
Use Tone Detector (TDET):	<input type="checkbox"/>
Time of Day Schedule (TOD):	0
Entry is a VNS Route (VNS):	<input type="checkbox"/>
Conversion to LDN (CNV):	<input type="checkbox"/>
Expensive Route (EXP):	<input type="checkbox"/>
Facility Restriction Level (FRL):	0 (0 - 7)
Digit Manipulation Index (DMI):	6
ISL D-Channel Down Digit Manipulation Index (ISDM):	0 (0 - 999)
Free Calling Area Screening Index (FCI):	0
Free Special Number Screening Index (FSNI):	0
Business Network Extension Route (BNE):	<input type="checkbox"/>
Strategy on Congestion (SBOC):	No Reroute (NRR)

Figure 46 – RLI_6 to tandem calls from PSTN to CS1000E_B

Create Digit Manipulation Block

Create DMI_13 for Incoming calls from CS1000E_B and Outgoing to PSTN; Figure 47

NORTEL CS 1000 ELEMENT MANAGER Help

Managing: 47.248.100.138 Username: admin
Dialing and Numbering Plans » [Electronic Switched Network \(ESN\)](#) » Customer 00 » Network Control & Services » [Digit Manipulation Block List](#) » Digit Manipulation Block

Digit Manipulation Block

Input Description	Input Value
Digit Manipulation Index numbers (DMI):	13
Number of leading digits to be Deleted (DEL):	0 (0 - 19)
Insert (INST):	91
IP Special Number (ISPN):	<input type="checkbox"/>
Call Type to be used by the manipulated digits (CTYP):	Unknown call type (UKWN)

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Figure 47 – DMI_13 for incoming from CS1000E_B and outgoing calls to PSTN

Create DMI_4 for incoming calls from CS1000E_B; Figure 48

The screenshot shows the Nortel CS 1000 Element Manager web interface. The left sidebar contains a navigation menu with categories like Home, Links, System, Customers, Routes and Trunks, Dialing and Numbering Plans, and Tools. The main content area is titled 'Digit Manipulation Block' and shows the configuration for a specific block. The breadcrumb trail indicates the path: Dialing and Numbering Plans » Electronic Switched Network (ESN) » Customer 00 » Network Control & Services » Digit Manipulation Block List » Digit Manipulation Block.

Input Description	Input Value
Digit Manipulation Index numbers (DMI):	4
Number of leading digits to be Deleted (DEL):	3 (0 - 19)
Insert (INST):	
IP Special Number (ISPN):	<input type="checkbox"/>
Call Type to be used by the manipulated digits (CTYP):	Call type will not be changed (NCHG)

At the bottom of the configuration area are buttons for Submit, Refresh, Delete, and Cancel. The footer of the page indicates 'Copyright © 2002-2007 Nortel Networks. All rights reserved.'

Figure 48 – DMI_4 to terminate calls from CS1000E_B

Create DMI_6: (Delete: 6) for incoming calls from PSTN and tandem calls to CS1000E_B; Figure 49

The screenshot shows the Nortel CS 1000 Element Manager web interface, similar to Figure 48 but for a different configuration. The breadcrumb trail is: Dialing and Numbering Plans » Electronic Switched Network (ESN) » Customer 00 » Network Control & Services » Digit Manipulation Block List » Digit Manipulation Block.

Input Description	Input Value
Digit Manipulation Index numbers (DMI):	6
Number of leading digits to be Deleted (DEL):	6 (0 - 19)
Insert (INST):	711
IP Special Number (ISPN):	<input type="checkbox"/>
Call Type to be used by the manipulated digits (CTYP):	Location Code (LOC)

At the bottom of the configuration area are buttons for Submit, Refresh, Delete, and Cancel. The footer of the page indicates 'Copyright © 2002-2009 Nortel Networks. All rights reserved.'

Figure 49 – DMI_6 for incoming calls from PSTN and tandem to CS1000E_B

Configure CS1000E 6.0 B:

1. Create IP Telephony Node on CS1000E

This section describes the steps for creating Node ID (2001) in CS 1000 network. Enter Element Manager through the IE browser (in IE address bar, type IP address of the Node IP or TLAN of Signaling Server).

- Input Node ID and press Add.
- Enter TLAN, ELAN IP address of Signaling Server.

Node 2001 was added to be configured as the SIP gateway to the Enterprise services, figure 50

The screenshot shows the 'CS 1000 ELEMENT MANAGER' interface. On the left is a navigation tree with categories like UCM Network Services, Home, Links, System, Alarms, Maintenance, Core Equipment, Loops, Superloops, MSDL/MISP Cards, Conference/TDS/Multifrequency, Tone Senders and Detectors, Peripheral Equipment, IP Network, Nodes, Servers, Media Cards, Maintenance and Reports, Media Gateways, Zones, Host and Route Tables, Network Address Translation, QoS Thresholds, Personal Directories, Unicode Name Directory, Interfaces, Engineered Values, Emergency Services, Geographic Redundancy, Software, Customers, and Routes and Trunks. The main content area is titled 'IP Telephony Nodes' and includes a sub-header 'System » IP Network » IP Telephony Nodes'. Below this is a prompt 'Click the Node ID to view or edit its properties.' and buttons for 'Add...', 'Import...', 'Export...', and 'Delete'. There are also 'Print' and 'Refresh' links. A table follows with the following data:

Node ID	Components	Enabled Applications	ELAN IP	TLAN IP	Status
2001	1	LTPS, Gateway (SIPGw)	-	192.168.10.11	Synchronized

At the bottom of the table, there are checkboxes for 'Show: Nodes' (checked) and 'Component Servers and Cards' (unchecked). The footer of the interface reads 'Copyright © 2002-2009 Nortel Networks. All rights reserved.'

Figure 50 – Node Configured as Enterprise Service SIP Gateway

The node IP information is added. For the primary proxy, Enter the IP address of the SIP Proxy Server (SPS). Use UDP port 5060 for SIP communication as shown in figure 51

Nortel CS 1000 ELEMENT MANAGER

System » IP Network » IP Telephony Nodes » Node Details

Node Details (ID: 2001 - LTPS, PD, Gateway (SIPGw))

Node ID: 2001 (0-9999)
 Call Server IP Address: 192.168.10.6
 Telephony LAN (TLAN) Node IP Address: 192.168.10.11
 Subnet Mask: 255.255.255.0
 Embedded LAN (ELAN) Gateway IP address: 192.168.100.1
 Subnet Mask: 255.255.255.0

IP Telephony Node Properties Applications (click to edit configuration)

- Voice Gateway (VGW) and Codecs
- Quality of Service (QoS)
- Terminal Proxy Server (TPS)

Associated Signaling Servers & Cards

Select to add	Add	Remove	Make Leader	Print	Refresh
<input type="checkbox"/> nd1-car1					

Note: Only server(s) that are not part of any other IP telephony node and deployed application(s) that match the service(s) selected for this node are available in the servers list.

Figure 51 – Node Details Configuration

Nortel CS 1000 ELEMENT MANAGER

System » IP Network » IP Telephony Nodes » Node Details » Virtual Trunk Gateway Configuration

Node ID: 2001 - Virtual Trunk Gateway Configuration Details

General | SIP Gateway Settings | SIP Gateway Services

Vtrk Gateway Application: ☒ Enable gateway service on this Node

General

Vtrk Gateway Application: SIP Gateway (SIPGw)
 SIP Domain name: interop.com
 Local SIP Port: 5060 (1 - 65535)
 Gateway endpoint name: car2_ss2
 Gateway password:
 Enable failsafe NRS: ☐

SIP Gateway Settings

TLS Security: Security Disabled
 Port: 5061 (1 - 65535)
 Number of Byte Re-negotiation: 0
 Options: ☐ Client Authentication
☐ X509 certificate authority

Proxy Or Redirect Server:

Primary TLAN IP Address: 192.168.10.60
 Port: 5060 (1 - 65535)
 Transport protocol: UDP
 Options: ☒ Support registration
☐ Primary CDS Proxy

Secondary TLAN IP Address: 0.0.0.0
 Port: 5060 (1 - 65535)
 Transport protocol: UDP
 Options: ☐ Support registration
☐ Secondary CDS Proxy

Virtual Trunk Network Health Monitor

☒ Monitor IP Addresses (listed below)
 Information will be captured for the IP addresses listed below.
 Monitor IP:
 Add
 Remove

* Required Value. Note: Changes made on this page will NOT be transmitted until the Node is also saved.

Save Cancel

Figure 52 – Trunk Gateway Configuration Details

In the signaling server properties, the Line TPS will be enabled if this signaling server will be used for IP set registration. If the role of this server is SIP gateway, only then this can be left unchecked.

2. Create D-channel (DCH)

- Launch Element Manager of CS 1000 6.0
- Choose D-Channels, enter D-channel number (i.e.: 101), select DCH for type

Click Add to create DCH 101, figure 53

NORTEL CS 1000 ELEMENT MANAGER

Managing: **192.168.10.6**
Routes and Trunks » D-Channels » D-Channels 101 Property Configuration

D-Channels 101 Property Configuration

- Basic Configuration

Input Description	Input Value
Action Device And Number (ADAN) (TYPE)	DCH
D channel Card Type (CTYP)	DQIP
Designator (DES)	Enterprise
Recovery to Primary (RCVP)	<input type="checkbox"/>
PRI loop number for Backup D-channel (BCHL)	
User (USR)	Integrated Services Signaling Link Dedicated (ISLD)
Interface type for D-channel (IFC)	Meridian Meridian1 (SL1)
Country (CNTY)	ETS 300 = 102 basic protocol (ETSI)
D-Channel PRI loop number (DCHL)	
Primary Rate Interface (PRI)	<input type="text"/> more PRI
Secondary PRI2 loops (PRI2)	<input type="text"/>
Meridian 1 node type (SIDE)	Slave to the controller (USR)
Release ID of the switch at the far end (RLS)	25
Central Office switch type (CO_TYPE)	100% compatible with Bellcore standard (STD)
Integrated Services Signaling Link Maximum (ISLM)	4000 Range: 1 - 4000
Signaling Server Resource Capacity (SSRC)	1800 Range: 0 - 4000

[+ Basic options \(BSCOPT\)](#)
[+ Advanced options \(ADVOPT\)](#)

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Figure 53 – D-Channel Property Configuration

Also click on Basic Options and edit the Remote Capabilities (RCAP). Enable MWI if CS1K hosted voice mail will be used.

3. Create Route

Create route 101 using DCH 101 for SIP trunks, figure 54

Managing: 192.168.10.6
Routes and Trunks > Routes and Trunks > Customer 0, Route 101 Property Configuration

Customer 0, Route 101 Property Configuration

- Basic Configuration

Input Description	Input Value
Route Data Block (RDB) (TYPE)	RDB
Customer number (CUST)	00
Route Number (ROUT)	101
Designator field for trunk (DES)	ENTERPRISE
Trunk Type (TKTP)	TIE
Incoming and Outgoing trunk (ICOG)	Incoming and Outgoing (IAO)
Access Code for the trunk route (ACOD)	8101
Trunk type M911P (M911P)	<input type="checkbox"/>
The route is for a virtual trunk route (VTRK)	<input checked="" type="checkbox"/>
Zone for codec selection and bandwidth management (ZONE)	255 Range: 0 - 255
Node ID of signaling server of this route (NODE)	2001 Range: 0 - 9999
Protocol ID for the route (PCID)	SIP (SIP)
Print Correlation ID in CDR for the route (CRID)	<input type="checkbox"/>
Integrated Services Digital Network option (ISDN)	<input checked="" type="checkbox"/>
Mode of operation (MODE)	Route uses ISDN Signaling Link (ISLD)
D channel number (DCH)	101 Range: 0 - 254
Interface type for route (IFC)	Meridian M1 (SL1)

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Figure 54 – Route Property Configuration

Configure Route 101 for SIP trunks, figure 55

Managing: 192.168.10.6
Routes and Trunks > Routes and Trunks > Customer 0, Route 101 Property Configuration

Customer 0, Route 101 Property Configuration

- Basic Configuration

Input Description	Input Value
D channel number (DCH)	101 Range: 0 - 254
Interface type for route (IFC)	Meridian M1 (SL1)
Private Network Identifier (PNI)	00001 Range: 0 - 32700
Network Calling Name Allowed (NCNA)	<input checked="" type="checkbox"/>
Network Call Redirection (NCRD)	<input checked="" type="checkbox"/>
Trunk Route Optimization (TRO)	<input type="checkbox"/>
Recognition of DTI2 ABCD FALT signal for ISL (FALT)	<input type="checkbox"/>
Channel Type (CHTY)	B-channel (BCH)
Call Type for outgoing direct dialed TIE route (CTYP)	Unknown Call type (UKWN)
Insert ESN Access Code (INAC)	<input checked="" type="checkbox"/>
Integrated Service Access Route (ISAR)	<input type="checkbox"/>
Display of Access Prefix on CLID (DAPC)	<input type="checkbox"/>
Mobile Extension Route (MBXR)	<input type="checkbox"/>

- Basic Route Options

Input Description	Input Value
Billing Number Required (BILN)	<input type="checkbox"/>
Call Detail Recording (CDR)	<input type="checkbox"/>
Controls or timers (CNTL)	<input type="checkbox"/>
Conventional (Tie trunk only) (CNVT)	<input type="checkbox"/>
Incoming DID Digit Conversion on this route (IDC)	<input type="checkbox"/>
Multifrequency Compelled or MFC Signaling (MFC)	No MFC (NO)
Process Notification Networked Calls (PNNC)	<input type="checkbox"/>

+ Network Options

+ General Options

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Figure 55 – Route Configuration

4. Create Trunk (figure 56)

CS 1000 ELEMENT MANAGER

Help | Logout

Home
Links
Virtual Terminals
Bookmarks
System
Alarms
Maintenance
Core Equipment
Peripheral Equipment
IP Network
Nodes: Servers, Media Cards
Maintenance and Reports
Media Gateways
Zones
Host and Route Tables
Network Address Translation
QoS Thresholds
Personal Directories
Interfaces
Engineered Values
Emergency Services
Software
Customers
Routes and Trunks
Routes and Trunks
D-Channels
Digital Trunk Interface
Dialing and Numbering Plans
Electronic Switched Network
Network Routing Service
Flexible Code Restriction
Incoming Digit Translation
Tools
Backup and Restore
Call Server Initialization
Date and Time

Managing: 192.168.10.6
Routes and Trunks » Routes and Trunks » Customer 0, Route 101, Trunk 1 Property Configuration

Customer 0, Route 101, Trunk 1 Property Configuration

- Basic Configuration

Input Description	Input Value
Trunk data block (TYPE)	PTI
Terminal Number (TN)	100 0 01 00
Designator field for trunk (DES)	ENTERPRISE
Extended Trunk (XTRK)	VTRK
Route number, Member number (RTMB)	101 1
Level 3 Signaling (SIGL)	
Card Density (CDEN)	80
Start arrangement Incoming (STRI)	Wink or Fast Flash (WINK)
Start arrangement Outgoing (STRO)	Wink or Fast Flash (WINK)
Trunk Group Access Restriction (TGAR)	0
Channel ID for this trunk, (CHID)	1
Increase or decrease the member numbers (INC)	Increase channel and member number (YES)
Class of Service (CLS)	Edit

+ Advanced Trunk Configurations

Save

Delete

Cancel

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Figure 56 – Trunk Property Configuration

Disable Media Security (sRTP) at the trunk level, figure 57

Managing: 192.168.10.6
Routes and Trunks » Routes and Trunks » Customer 0, Route 101, Trunk 1 Property Configuration » Class of Service Configuration

Class of Service Configuration

Input Description	Input Value
- ACD Priority (CLS)	ACD Priority not required (APN)
- Analog Semi-Permanent Connections (CLS)	Analog Semi-Permanent Connections Denied (SPCD)
- ARF Supervised COT (CLS)	
- Barring (CLS)	
- Battery Supervised COT (CLS)	
- Busy Tone Supervised COT (CLS)	
- Calling Line Identification (CLS)	
- Calling party (CLS)	Calling party Denied (CND)
- Central Office Ringback (CLS)	
- Centrex Switchhook Flash (CLS)	Centrex Switchhook Flash Denied (THFD)
- Dial Pulse (CLS)	Digitone (DTN)
- DTR PAD value (CLS)	
- Echo Canceling (CLS)	Echo Canceling Denied (ECD)
- Hong Kong DTI (CLS)	
- Loop Break Supervised COT (CLS)	
- Make-break ratio for dial pulse (CLS)	10 pulses per second (P10)
- Manual Incoming (CLS)	
- Media Security (CLS)	Media Security Never (MSNV)
- Network Hook Flash Over M911P (CLS)	

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Figure 57 – Class of Service Configuration

5. Create Dialing Plan

Create Location Code

Create LOC 511 (Use RLI_5) for outgoing calls to CS1000E_A; Figure 58

Figure 58 – LOC_511 for out going calls to CS1000E_A

Create Numbering Plan Area Code

Create NPA_613 (RLI_70) for Outgoing calls to PSTN through CS1000E_A; Figure 59

Figure 59 – NPA_613 for Outgoing calls to PSTN

Create Home Location Code

Create HLOC_521 (Use DMI_4) for incoming calls from CS1000E_A; Figure 60

NORTEL CS 1000 ELEMENT MANAGER [Help](#)

Dialing and Numbering Plans » [Electronic Switched Network \(ESN\)](#) » Customer 00 » Numbering Plan (NET) » Access Code 2 » [Home Location Code List](#) » Home Location Code

Home Location Code

Input Description	Input Value
Home Location code (HLOC):	<input type="text" value="521"/>
Digit Manipulation Index (DMI):	<input type="text" value="4"/>

- UCM Network Services
- Home
- Links
 - Virtual Terminals
- System
 - + Alarms
 - Maintenance
 - + Core Equipment
 - Peripheral Equipment
 - + IP Network
 - + Interfaces
 - Engineered Values
 - + Emergency Services
 - + Geographic Redundancy
 - + Software
- Customers
- Routes and Trunks
 - Routes and Trunks
 - D-Channels
 - Digital Trunk Interface
- Dialing and Numbering Plans
 - [Electronic Switched Network](#)
 - Flexible Code Restriction
 - Incoming Digit Translation
- Phones
 - Templates
 - Reports
 - Properties
 - Migration
- Tools
 - + Backup and Restore
 - Date and Time
 - + Logs and reports
- Security
 - + Policies
 - + Login Options

Figure 60 – Create HLOC 521 for incoming calls from CS1000E_A

Create Route List Block

Create RLI_5 for outgoing calls to CS1000E_A (Use DMI_0)

CS 1000 ELEMENT MANAGER
Help

- UCM Network Services
- Home
- Links
- Virtual Terminals
- System
 - + Alarms
 - Maintenance
 - + Core Equipment
 - Peripheral Equipment
 - + IP Network
 - + Interfaces
 - Engineered Values
 - + Emergency Services
 - + Geographic Redundancy
 - + Software
- Customers
- Routes and Trunks
 - Routes and Trunks
 - D-Channels
 - Digital Trunk Interface
- Dialing and Numbering Plans
 - **Electronic Switched Network**
 - Flexible Code Restriction
 - Incoming Digit Translation
- Phones
 - Templates
 - Reports
 - Properties
 - Migration
- Tools
 - + Backup and Restore
 - Date and Time
 - + Logs and reports
- Security
 - + Policies
 - + Login Options

Data Entry of a Route List Block

Route List Block Index: 5

Input Description	Input Value
Entry Number for the Route List (ENTR):	0
Local Termination entry (LTER):	<input type="checkbox"/>
Route Number (ROUT):	101
Skip Conventional Signaling (SCNV):	<input type="checkbox"/>
Use Tone Detector (TDET):	<input type="checkbox"/>
Time of Day Schedule (TOD):	0
Entry is a VNS Route (VNS):	<input type="checkbox"/>
Conversion to LDN (CNV):	<input type="checkbox"/>
Expensive Route (EXP):	<input type="checkbox"/>
Facility Restriction Level (FRL):	(0 - 7)
Digit Manipulation Index (DMI):	0
ISL D-Channel Down Digit Manipulation Index (ISDM):	(0 - 999)
Free Calling Area Screening Index (FCI):	0
Free Special Number Screening Index (FSNI):	0
Business Network Extension Route (BNE):	<input type="checkbox"/>
Strategy on Congestion (SBOC):	No Reroute (NRR)
- QSIG Alternate Routing Causes (COPT):	QSIG Alternate Routing Cause 1
ISDN Drop Back Busy (IDBB):	Drop Back Disabled (DBD)

Figure 61 – RLI_5 (use DMI_0) for Outgoing calls to CS1000E_A

CS 1000 ELEMENT MANAGER
[Help](#)

- UCM Network Services
- Home
- Links
 - Virtual Terminals
- System
 - + Alarms
 - + Maintenance
 - + Core Equipment
 - + Peripheral Equipment
 - + IP Network
 - + Interfaces
 - Engineered Values
 - + Emergency Services
 - + Geographic Redundancy
 - + Software
- Customers
- Routes and Trunks
 - Routes and Trunks
 - D-Channels
 - Digital Trunk Interface
- Dialing and Numbering Plans
 - **Electronic Switched Network**
 - Flexible Code Restriction
 - Incoming Digit Translation
- Phones
 - Templates
 - Reports
 - Properties
 - Migration
- Tools
 - + Backup and Restore
 - Date and Time
 - + Logs and reports
- Security
 - + Policies
 - + Login Options

Dialing and Numbering Plans » [Electronic Switched Network \(ESN\)](#) » Customer 00 » Network Control & Services » [Digit Manipulation Block List](#) » Digit Manipulation Block

Digit Manipulation Block

Input Description	Input Value
Digit Manipulation Index numbers (DMI):	<input type="text" value="4"/>
Number of leading digits to be Deleted (DEL):	<input type="text" value="3"/> (0 - 19)
Insert (INST):	<input type="text"/>
IP Special Number (ISPN):	<input type="checkbox"/>
Call Type to be used by the manipulated digits (CTYP):	<input type="text" value="Location Code (LOC)"/>

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Figure 62 – DMI_4 for incoming call to this CS1000E_B

Configure SIP Proxy Server (SPS)

Create gateway endpoints on SPS

NORTEL NETWORK ROUTING SERVICE MANAGER Help | Logout

«Common Manager

- System
 - NRS Server
 - Database
 - System Wide Settings
- Numbering Plans
 - Domains
 - Endpoints
 - Routes
 - Network Post-Translation
 - Collaborative Servers
- Tools
 - SIP Phone Context
 - Routing Tests
 - H.323
 - SIP
 - Backup
 - Restore
 - GK/NRS Data upgrade

Managing: ☒ Active database **192.168.10.60**
☐ Standby database [Numbering Plan > Endpoints](#)

Search for Endpoints Hide

Enter an endpoint ID (use * for all) and click Search. You may narrow the search by specifying a particular domain.

Endpoint ID:

Limit results to Domain: / /

Results per page:

Gateway Endpoints (5) **User Endpoints (0)**

SIP phone context... Refresh

ID	Supported Protocols	Call Signaling IP	Description	# of Routing Entries	Context
1 <input type="checkbox"/> BCM50r2	Static SIP endpoint / NCS	47.248.100.215	BCM50r2	2	interop.com / udp / cdp
2 <input type="checkbox"/> QCS-MCM	Dynamic SIP endpoint / NCS	47.248.100.123	OCS	1	interop.com / udp / cdp
3 <input type="checkbox"/> car1_ss2	Dynamic SIP endpoint / NCS	192.168.10.12	car1_ss2	1	interop.com / udp / cdp
4 <input type="checkbox"/> car2_ss2	Dynamic SIP endpoint / NCS	192.168.10.11	car2_ss2	2	interop.com / udp / cdp
5 <input type="checkbox"/> mn118_mcs_usr	Dynamic SIP endpoint / NCS	Not registered	hnh	1	interop.com / udp / cdp

1 - 5 of 5 Gateway Endpoint(s) Page 1 of 1 First| Previous| Next| Last

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Figure 623 – SIP Gateway Endpoint Creation

Create routing entries for each of gateway endpoints on SPS, figure 64

NORTEL NETWORK ROUTING SERVICE MANAGER Help | Logout

«Common Manager

- System
 - NRS Server
 - Database
 - System Wide Settings
- Numbering Plans
 - Domains
 - Endpoints
 - Routes
 - Network Post-Translation
 - Collaborative Servers
- Tools
 - SIP Phone Context
 - Routing Tests
 - H.323
 - SIP
 - Backup
 - Restore
 - GK/NRS Data upgrade

Managing: ☒ Active database **192.168.10.60**
☐ Standby database [Numbering Plan > Routes](#)

Search for Routing Entries

Enter a DN Prefix and DN Type (use * for all) and click Search. You may narrow the search by specifying a particular domain.

DN Prefix: DN Type:

Limit results to Domain: / /

Endpoint Name:

Results per page:

Routing Entries (7) **Default Routes (0)**

Routing test... Refresh

DN Prefix	DN Type	Route Cost	SIP URI Phone Context	Context
3 <input type="checkbox"/> 521	Private level 1 regional (UDP location code)	1	udp	interop.com / udp / cdp / car2_ss2
4 <input type="checkbox"/> 613	Private level 1 regional (UDP location code)	1	udp	interop.com / udp / cdp / car2_ss2
5 <input type="checkbox"/> 511	Private level 1 regional (UDP location code)	1	udp	interop.com / udp / cdp / car1_ss2

1 - 7 of 7 Routing Entry(ies) Page 1 of 1 First| Previous| Next| Last

Figure 634 – Routing Entries for Gateway Endpoints

CS1000E SIPLINE CONFIGURATION

In this section, it shows how to configure a SIP LINE system on CS1000E. Follow the bellow steps to set up the SIP LINE server.

Configure SIP LINE CS1000E in Element Manager

Figure 64 show hot to add SIP LINE Node 1002 under System -> IP Network -> IP Telephony Nodes

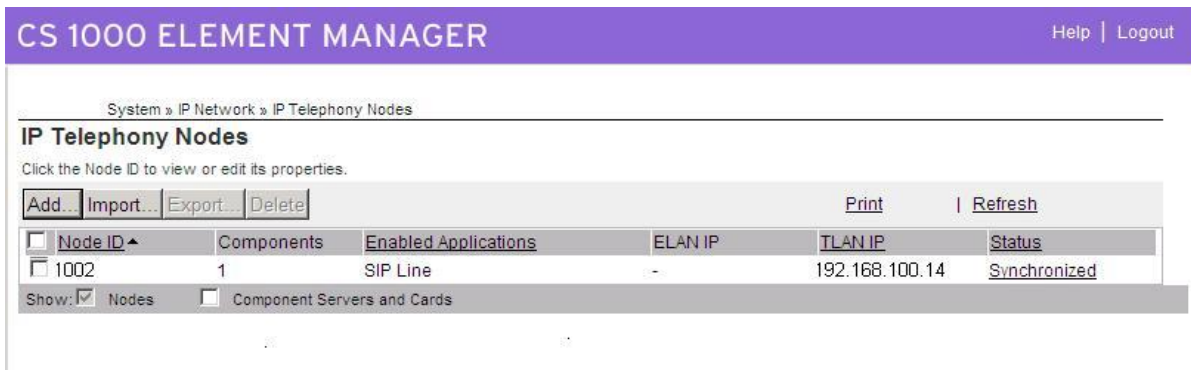


Figure 645 – IP Telephony Nodes

Figure 66, 67 and 68 show how to set up the SIP LINE Node 1002 configuration details
SAVE and SYNC are required – And then APPSTART RESTART on SLG server.

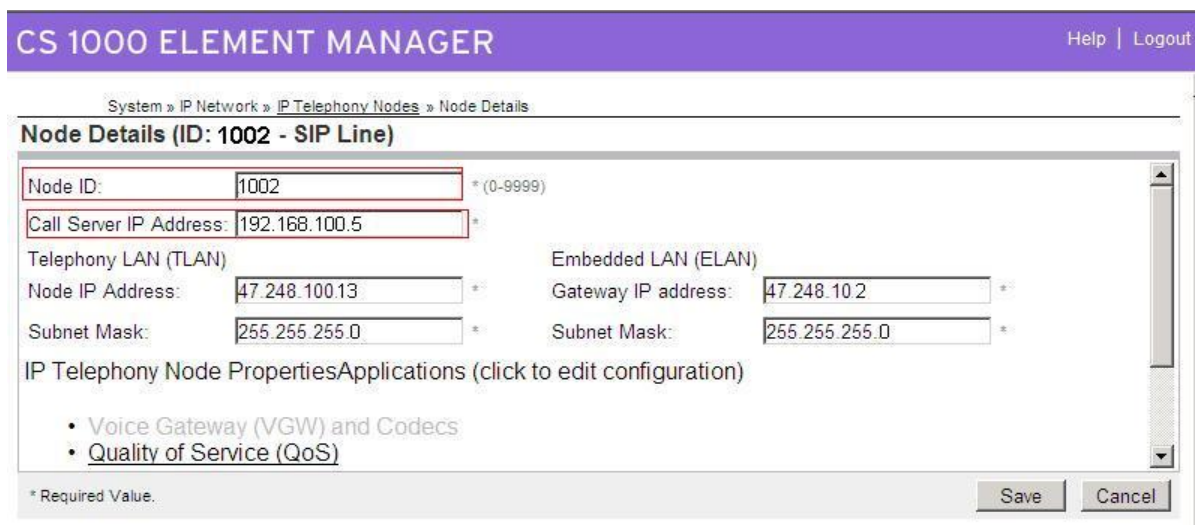


Figure 656 – Node Configuration Details

CS 1000 ELEMENT MANAGER Help | Logout

System » IP Network » IP Telephony Nodes » Node Details » SIP Line Configuration

Node ID: 1002 - SIP Line Configuration Details

General | SIP Line Gateway Settings | SIP Line Gateway Service

SIP Line Gateway Application: ☒ Enable gateway service on this Node

General

SIP Domain name: *

SLG endpoint name:

SLG Group ID:

SLG Local Sip Port: (1 - 65535)

SLG Local Tls Port: (1 - 65535)

SIP Line Gateway Settings

Security Policy:

Number of Byte Re-negotiation:

Options: ☐ Client Authentication
☐ x509 Certificate Authentication Enabled

SIP Line Gateway Service

Branch / GR Office Settings:

Virtual Trunk Network Health Monitor

☐ Monitor IP Addresses (listed below)
Information will be captured for the IP addresses listed below.

Monitor IP:

Monitor addresses:

* Required Value. Note: Changes made on this page will NOT be transmitted until the Node is also saved.

Figure 667 – Node Configuration Details (Cont...)

CS 1000 ELEMENT MANAGER Help | Logout

System » IP Network » IP Telephony Nodes » Node Details » SIP Line Configuration

Node ID: 1002 - SIP Line Configuration Details

Branch / GR Office Settings:

SLG Role:

SLG Mode:

MO SLG IP:

MO SLG Port: (1 - 65535)

MO SLG Transport:

GR SLG IP:

GR SLG Port: (1 - 65535)

GR SLG Transport:

* Required Value. Note: Changes made on this page will NOT be transmitted until the Node is also saved.

Figure 678 – Node Configuration Details (Cont...)

Configure CS1000E Call Server

For the configuration of SIP Line on Call Server, one needs to use command line to set it up. Follow the bellow steps to accomplish that.

Packages Required for SIP line on Call Server of CS1000E, these are keycode enablement

1. SLS_Package – 417 - SIP Line Service
2. FFC- 139 - Flexible Feature Codes
3. SIP_LINE_NT_PKG – 415 - Avaya SIP Line Package
4. SIP_LINE_3P_PKG – 416 - 3rdParty SIP Line Package

Configure SIPL service in LD15

LD 15
REQ CHG
TYPE SLS
CUST 0
SIPL_ON YES
SIPD **INTEROP.COM**
UAPR **222** - DN prefix used to auto-generate UADN for all SIPL clients of this customer
NMME NO

Configure DCH for SIPL in LD 17

LD 17
REQ CHG
TYPE ADAN
ADAN new dch 11
ADAN DCH 11
CTYP **DCIP**
DES **SIPL**
USR **ISLD**
ISLM 4000
SSRC 1800
OTBF 32
NASA NO
IFC **SL1**
CNEG 1
RLS ID 25
RCAP
MBGA NO
H323
OVLN NO
OVLS NO

Configure ELAN AML link in LD 17

LD 17
REQ CHG
TYPE ADAN
ADAN new elan 32
ADAN ELAN **32** – new AML ELAN link, link number should be bigger or equal to 32
CTYP **ELAN**
DES **SIPL**
N1 512

Configure VAS ID for AML link in LD 17

LD 17
REQ CHG

TYPE VAS
VAS new
VSID **32** – VAS ID number
ELAN **32** – Defined in step 3

Configure SIPL route

LD 16
REQ new
TYPE rdb
CUST 0
ROUTE 11
DES **SIPL**
TKTP **TIE**
...
VTRK **YES**
ZONE **10** – virtual trunk zone defined in LD117
PCID **SIPL**
...
NODE **1002** – node ID of SIPL node
DTRK NO
ISDN YES
MODE **ISLD**
DCH **11** – DCH defined in step 2
IFC **SL1**
PNI **00001**
NCNA **YES**
NCRD **YES**
TRO NO
FALT NO
CTYP UKWN
INAC **YES**
ISAR NO
DAPC NO
...
ICOG **IAO**
...
ACOD **8011** – route access code

Configure SIPL trunks

LD 14
REQ **NEW 256** – e.g. create 256 trunks
TYPE **IPTI**
TN **124 0 0 0** - starting TN for virtual trunks
DES **SIPL**
CUST 0
RTMB **11 1** – route number and member
CHID 1
TGAR **0**
STRI **IMM**
STRO **IMM**
CLS **UNR**

Check status of the details configuration SIPL link is up on Call Server and SIP line Gateway

On Call Server

```
>*ld 96
DCH 011 : OPER    EST  ACTV AUTO    DES : SIPL_N1402
```

On SLG

```
[nortel@vrf14-sls ~]$ slgShow
=== VTRK ===
```

```
===== General =====
```

```
SLG State      = AppReady
Total User Registered = 1
```

```
===== AML Info =====
```

hAppBlk	TaskName	Tid	LinkState	NumRetry	LinkNum	Trace
0x1226c80	SLG	0xfb00	Up	0	32	0

Configure SIP Line Client

Setting password length for SIP line client using LD15

```
LD 15
REQ CHG
TYPE: FFC
TYPE FFC_DATA
CUST 0
```

SCPL **4** – password length is 4

Configure UEXT for SIPL client

```
LD 11
REQ NEW
TYPE UEXT
```

```
TN 104 0 00 11 - Virtual TN for SIPL client
CUST 0
UXTY SIPL – UEXT type must be SIPL
MCCL YES
SIPN 1
SIP3 1
FMCL 0
TLSV 0
```

**** Begin Note:**

Sigma phone: SIPN-SIP3-FMCL-TLSV = 1-0-0-0
SMC3456: SIPN-SIP3-FMCL-TLSV = 1-0-0-0

SipToneV: SIPN-SIP3-FMCL-TLSV = 0-1-0-0

***End Note

SIPU **4861** – SIPL userID, often set equal to DN of the phone

NDID **1002** – NodeID of the SIPL node

ZONE 001 – MO zone configured in LD 117

TGAR **0** – should be 0, if not we can dial to SipToneV

...

SCPW **1234** – password for SIPL client to login

...

CLS **UNR**

...

KEY 00 SCR **4861** – DN of the phone

CPND **NEW** – in case you want to set CLID for phone

NAME **set4861**

XPLN 20

DISPLAY_FMT FIRST, LAST

01 HOT U **2224861** – autogenerate when you enter information for KEY 0

Check current status set registration on SLG

```
[nortel@vrf14-sls ~]$ slgSetShowAll
```

```
=== VTRK ===
```

UserID	TN	Clients	Calls	SetHandle
4861	104-00-00-11	1	0	0xb7d8a0c8

SMC3456 Softphone

Link to download:

<http://livelink-ott.ca.nortel.com/livelink/livelink.exe?func=ll&objId=34471954&objAction=browse&sort=name&viewType=1>.

After installation on the PC and apply the Licence key which is required for activate the SMC to be used. Run the SMC3456, you will see figure

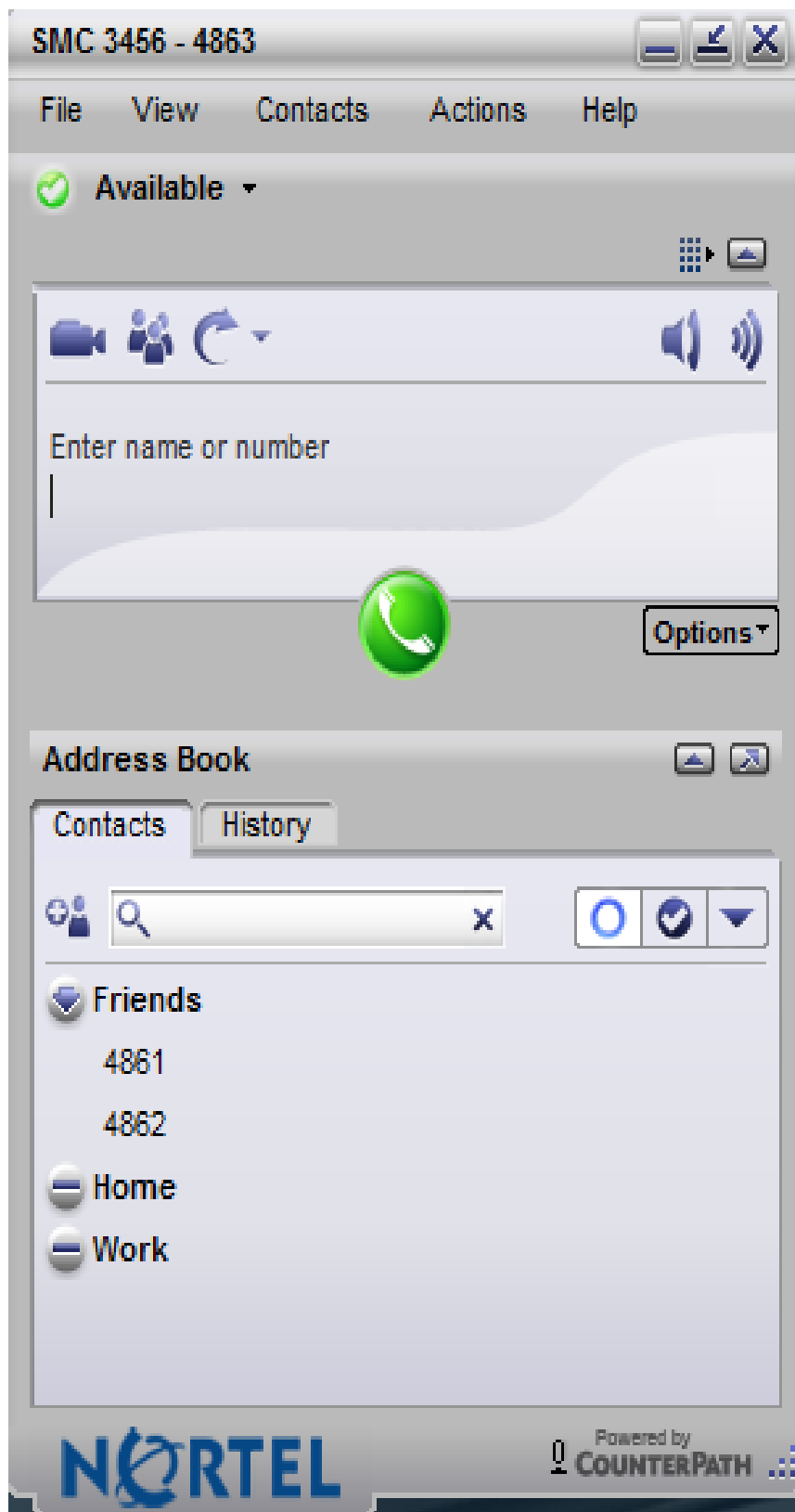


Figure 69 SMC Client

On the top menu bar, go to FILE -> PREFERENCES -> ADVANCED -> LOGIN SERVER
 ➔ No login server available

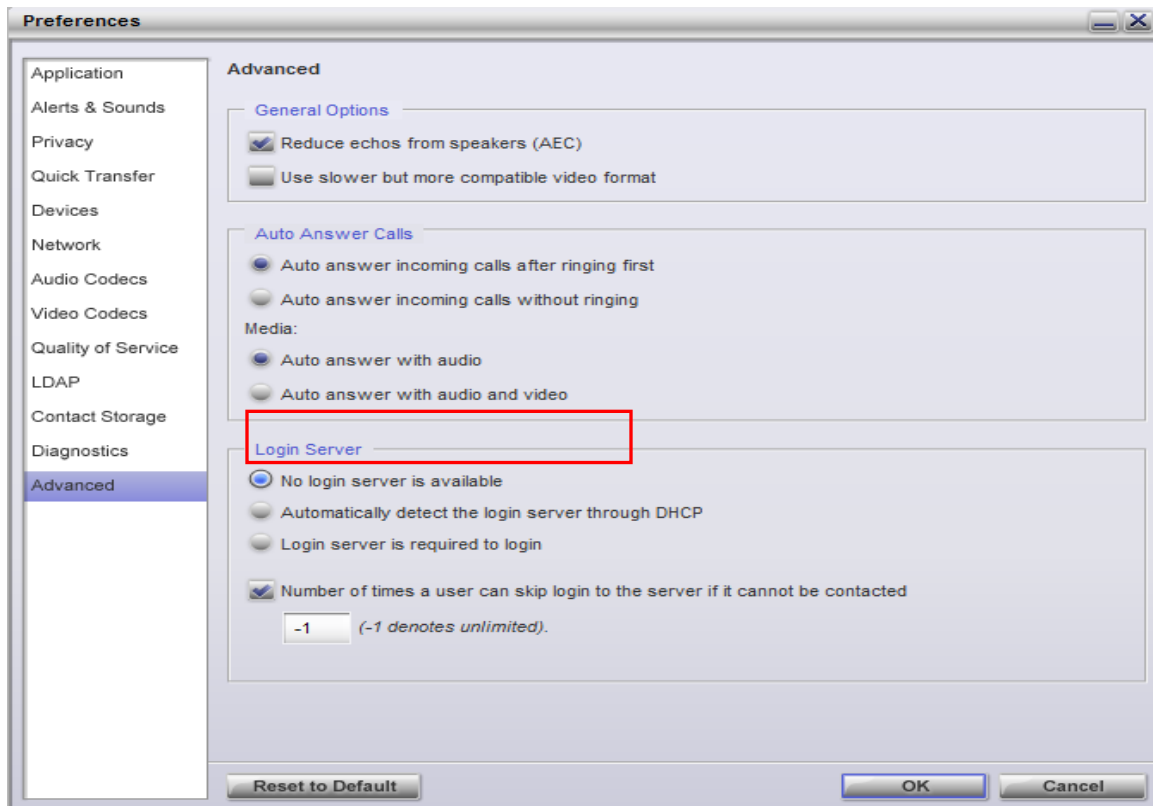


Figure 70 – Advanced Options Menu

Add a SIP Account on SMC3456

In order to create a SIP account for SMC3456 to be able to register to CS1000E SIP line server, From the top menu bar go to FILE -> ACCOUNT SETTINGS -> Add New SIP Account, see figure 70.

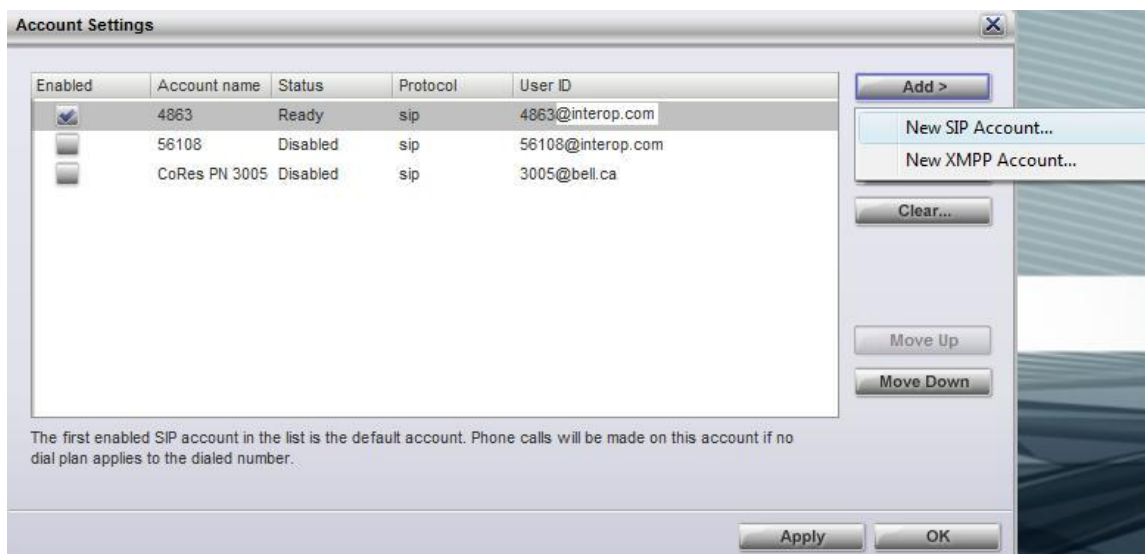


Figure 681 – Accounting Settings

The created account is appeared as figure 72.

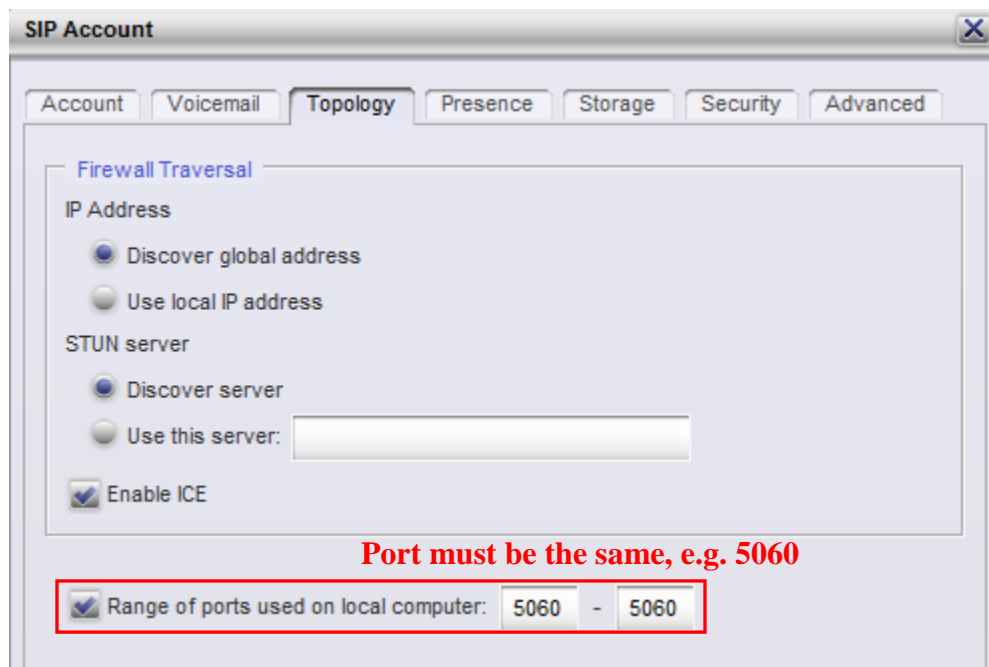


Figure 692 – Topology SIP Account Settings

Figure 72 shows how to set SIP account details by clicking on the Account menu tab.

The screenshot shows a window titled "SIP Account" with a close button in the top right corner. Below the title bar are several tabs: "Account", "Voicemail", "Topology", "Presence", "Storage", "Security", and "Advanced". The "Account" tab is currently selected. The form contains the following fields and options:

- Account name:** 4863
- Protocol:** SIP
- User Details section:**
 - User ID:** 4863@interop.com (highlighted with a red box). A placeholder text "e.g. joseph@domain.com" is visible to the right.
 - Password:** **** (highlighted with a red box).
 - Display name:** 4863
 - Authorization name:** 4863
- Domain Proxy section:**
 - ☒ Register with domain and receive calls
 - Send outbound via:**
 - ☐ Domain
 - ☒ Proxy Address: 192.168.100.13:5070 (highlighted with a red box)
- Dial plan:** #1\va.T;match=1;prestrip=2;

Figure 703 – SIP Account Details Setting

Figure 73 shows the newly created SIP account

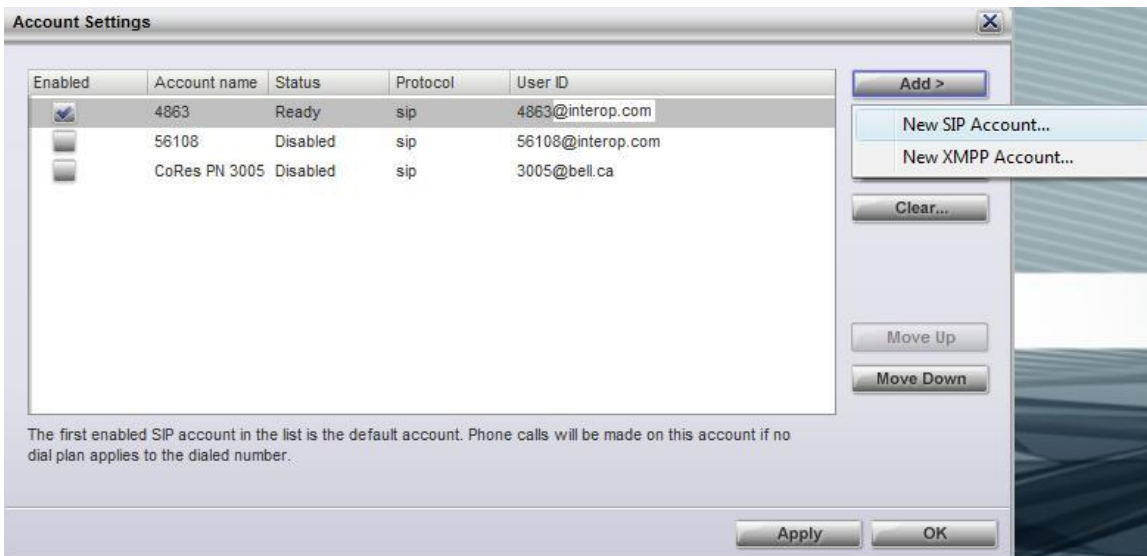


Figure 714 – Newly Created SIP Account

Provisioning SIP Phone Sets 1140 and 1120 on CS1000E Call Server

On CS1000E Call Server, use Command Line interface to configure the 1120 and 1140 phone sets.

```
TN 104 0 0 0
UXTY
DATE
PAGE
DES

DES SIPL
TN 104 0 00 00 VIRTUAL
TYPE UEXT
CDEN 8D
CTYP XDLC
CUST 0
UXTY SIPL
MCCL YES
SIPN 1
SIP3 0
SIPU 4861
NDID 1002
SUPR NO
SUBR DFLT MWI RGA CWI MSB
UXID
NUID
NHTN
CFG_ZONE 010
```

CUR_ZONE 010
 ERL 0
 ECL 0
 FDN
 TGAR 0
 LDN NO
 NCOS 7
 SGRP 0
 RNPG 0
 SCI 0
 SSU
 XLST
 SCPW 1234
 SFLT NO
 CAC_MFC 0
 CLS UNR FBD WTA LPR MTD FND HTD TDD HFD CRPD
 MWD LMPN RMMD SMWD AAD IMD XHD IRD NID OLD VCE DRG1
 POD DSX VMD SLKD CCSD SWD LND CNDD
 CFTD SFD MRD DDV CNID CDCA MSID DAPA BFED RCBF
 ICDD CDMD LLCN MCTD CLBD AUTU
 GPUD DPUD DNDD CFXD ARHD CLTD ASCD
 CPFA CPTA ABDD CFHD FICD NAID BUZZ AGRD MOAD
 UDI RCC HBTD AHA IPND DDGA NAMA MIND PRSD NRWD NRCD NROD
 DRDD EXR0
 USMD USRD ULAD CCBF RTDD RBDD RBHD PGND OCBF FLXD FTTC DNDY DNO3 MCBN
 FDSD NOVD VOLA VOUD CDMR ICRD MCDD T87D MSNV FRA PKCH
 CPND_LANG ENG
 HUNT
 PLEV 02
 DANI NO
 AST
 IAPG 0
 AACS NO
 ITNA NO
 DGRP
 MLWU_LANG 0
 MLNG ENG
 DNDR 0
 KEY 00 SCR 496855 0 MARP
 01 HOT U 2224861 MARP 0

Frontier Communication System configuration

Frontier will have to provide this configuration notes.

General Test Approach and Test Results

The focus of this interoperability compliance testing was to verify the SIP trunk connectivity between the Frontier Communication systems and Avaya Communication Server 1000E Release 6.0. The testing verified if the SIP signaling and media of the basic telephony features were communicating correctly. The following features were covered; basic calls, busy, music on hold, blind and consultative transfers, DTMF, MWI, codec negotiations, conference.

General Test Approach

The general test approach was to have Frontier Genband system connected to CS1000E SIP Gateway using Genband IP address. The SIP trunk communication should be established between CS1000E and Frontier Genband system. Calls can be made from end to end, i.e. PSTN phone can call through created route from Frontier Communication system to CS1000Es' analog, IP, SIP phones via SIP trunk. The main objectives were to verify the basic SIP trunk features:

- Basic call from PSTN phone to CS1000E phones
- Perform basic call operation: DTMF transmission, voicemail with MWI notification, busy, hold.
- Redirect call between users/clients/endpoints: blind/consultative transfers, call forward all call, busy and no answer.
- Perform codec negotiation
- Perform conferencing

Test Results

The objectives outlined in section 6.1 were verified and met. The following observations were made during the compliance testing:

- Dial to telephone number which begins with “*”, i.e. *xxxxx does not match required format on Avaya CS1000E.
- CPND, Call Party Name Display, does not support on test set up. Telephone number is displayed instead.
- Music on hold is not enabled on CS1000E. i.e. User won't hear music when call is put on hold
- Media Security is not enabled (MSEC = off in LD 17) on both SIP trunk and phones on CS1K locally
- PSTN1 calls CS1000E_PHONE. Call is established between PSTN1 and CS1000E_PHONE. CS1000E_PHONE does blind transfer or consult transfer to PSTN2 but PSTN2 does not ring. CS1000E_PHONE hears greeting “Sorry your long distant service is not established. Please contact...” instead of ringback tone. In order to resolve this, the work around can be used by setting of DORG (Display Originator) parameter in CS1K call server to “No”. However, this will cause a side effect on the CLID/CPND of PSTN phones. None of the CLID/CPND PSTN phones are displayed correct. This issue is due to the usage of proprietary MCDN feature of the CS1K to pass

the CLID information over the trunk, which does not understand by the Carrier Service Provider. Please refer to CR Q02141122.

- PSTN1 calls CS1000E_PHONE. Call is established between PSTN1 and CS1000E_PHONE. CS1000E_PHONE initiates a conference with PSTN2 but PSTN2 does not provide ringing tone. CS1000E hears greeting “Sorry your long distant service is not established. Please contact...” instead of ringback tone. Please refer to CR Q02135507. This issue can be resolved by configuration change in CS1K where DORG is set up to “No” in LD 86 as above.
- PSTN1 calls CS1000E_PHONE1. Call is established between PSTN1 and CS1000E_PHONE1 successfully. CS1000E_PHONE1 does blind transfer to CS1000E_PHONE2 which is set up to call forward no answer to PSTN2. The expectation is as follow: CS1000E_PHONE1 call is released, PSTN1 hears ring-back tone, CS1000E_PHONE2 rings for three times and then the call is forwarded to PSTN2. The actual result is as follow: CS1000E_PHONE2 is released after call is forwarded no answer to PSTN2. PSTN2 rings but PSTN1 does not hear ring back tone from PSTN2, and when PSTN2 answers the call, the call is dropped right away. This issue has been resolved by changing CS1K configuration where DORG is set up to “No” in LD 86. However, the CLID information was not displayed correctly on PSTN1.
- CS1000E_TDM calls CS1000E_IP_PHONE which is set up to call forward no answer to PSTN. User expects to hear PSTN phone ringing after CS1000E_IP_PHONE rings three times. Then the call should be forwarded to PSNT phone and the CS1000E_IP_PHONE is released so call is established between CS1000E_TDM and PSTN. However, CS1000E_TDM did hear the 3 ring-back tones, a silence period, then ring back tones and so on.... Please refer to CR Q02138017. Call will work if Media security (MSEC) is disabled for both SIP trunk and phones on CS1K locally. This CR is closed “No fix planned” with recommend MSEC needs to be set to “Off” in LD 17.
- PSTN1 calls to CS1000E_PHONE. Call is established successfully when CS1000E_PHONE answers. CS1000E_PHONE does blind transfer or consult transfer to PSTN2. CLID/CPND of PSTN1 and PSTN2 display CS1000E_PHONE instead of PSTN2 number is displayed on PSTN1 and PSTN1 number is displayed on PSTN2. Please refer CR Q02141122.
- CLID/CPND of PSTN2 displays incorrectly with the following scenarios:
PSTN1----call---CS1000_TDM---CFNA---PSTN2 displays "1771640-3" instead of PSTN1 number
PSTN1 ----call-----CS1000_IP_PHONE2-----CFAC-----PSTN2 displays "1771640-1" instead of PSTN1 number
PSTN1----call-----CS1000_IP_PHONE2-----CFB-----PSTN2 displays "1771640-3" instead of PSTN1 number

Please refer to CR Q02138314 as same root cause as Q02138314.

- FAX G711 Modem Pass Through Mode is not supported on Frontier network. Test cases from 5.4.3.1 to 5.4.3.4 were not tested.

- SIPLINE call failed on SU17. Issue on CS1K SIP line 6.0. Please refer to CR Q02129692

PSTN calls CS1000E_SIPLINE. There are two actual results when SIPLINE answers call as below:

1) SIPLINE rings and PSTN phone gets ringback tone. If SIPLINE answers the call, SIPLINE will display “Remote hold” without speech path between PSTN and SIPLINE.

2) Call is established with two ways speech path between SIPLINE and PSTN. However, SIPLINE is not released when PSTN ends call.

Addition, Simple call is failed between SIPLINE1 and SIPLINE2. Call is established between SIPLINE1 and SIPLINE2 successfully. However, SIPLINE2 is not released when SIPLINE1 ends call.

Simple call also failed between CS1000E_Unistim calls CS1000E_SIPLINE.

CS1000E_Unistim calls CS1000E_SIPLINE. Call is dropped right after CS1000E_SIPLINE answers call.

Verification Steps

This section includes some steps that can be followed to verify the solution is working.

Verify that calls are established with two-way voice path when making calls from one CS1000E phone to another on the local CS1000E.

Verify that IP phones, digital, analog (Fax) register successfully show as below:

Verify status of IP phone registered

```
[nortel@nd1-car1 ~]$ isetShow
```

```
=== TPS ===
```

1. Set Information

IP Address	NAT	Model Name	Type	RegType	State	Regd-TN	FWVsn
47.248.101.117		IP Phone 1120E		1120	Regular online	096-00-01-24	
C60							
47.248.101.120		IP Phone 2002 Phase 2		2002P2	Regular online	096-00-01-06	
DCJ							
47.248.101.116		IP Phone 1140E		1140	Regular online	096-00-01-26	C60
47.248.101.115		IP Phone 1220		1220	Regular online	096-00-01-	
05 C6O							

Verify status of digital phone registred:

```
LD 32
Stat 4 0 7
>ld 32
.stat 4 0 7
00 = UNIT 00 = IDLE (3904)
01 = UNIT 01 = IDLE (3902)
```

.....

Verify status of Analog (Fax machine registered):

```
LD 32
.stat 4 0 8
00 = UNIT 00 = IDLE (L500)
01 = UNIT 01 = IDLE (L500)
```

Verify the following basic calls in local CS1000E:

```
IP phone-----call-----IP phone
IP phone -----call-----SIP Line Client
IP Phone -----call-----Analog/Fax phone
IP Phone -----call-----Digital phone
SIP Line Client-----call-----Analog/Fax phone
SIP Line Client-----call-----Digital Phone
Analog/Fax phone-----call-----Digital Phone
User can verify the same for calls from oposite direction.
```

Verify that calls are established with two-way voice path and busy status under CS1000E call server as below:

Verify status of IP phones which are busy

```
[nortel@nd1-car1 ~]$ isetShow
=== TPS ===
```

Set Information

```
-----
  IP Address   NAT  Model Name      Type RegType  State      Regd-TN
UNIStimVsn
-----
47.248.101.117   IP Phone 1120E      1120  Regular busy  096-00-01-24  C6O
47.248.101.120   IP Phone 2002 Phase 2    2002P2 Regular busy  096-00-01-06 DCJ
47.248.101.116   IP Phone 1140E      1140  Regular busy  096-00-01-26  C6O
```

47.248.101.115 IP Phone 1220 1220 Regular busy 096-00-01-05 C60

Verify status of digital phone is busy

LD 32 .stat 4 0 7 000 = UNIT 00 = BUSY (3904)
01 = UNIT 01 = BUSY (3902)

.....

Verify status analog phone is busy

LD 32
.stat 4 0 8
00 = UNIT 00 = BUSY (L500)
01 = UNIT 01 = BUSY (L500)

Verify status of voice gateway if calls are established between IP phone/SIP line Clients to Analog/Digital phones or call to voice message

>>ld 32
NPR000
.stat 4 0 11
00 = UNIT 00 = BUSY (TRK)(IPTN REG)
01 = UNIT 01 = BUSY (TRK)(IPTN REG)
02 = UNIT 02 = BUSY (TRK)(IPTN REG)
03 = UNIT 03 = BUSY (TRK)(IPTN REG)

During the call, use pcap tool (ethereal/wireshark) at the TLAN media gateway card, RTP streams are going for call relate to analog, digital or voice message.

Verify that calls are established with two-way voice path when making calls from PSTN phone to Avaya phones on the CS1000 through Frontier Communication system via configured SIP trunk.

- Verify basic call between PSTN phones and Avaya phones. During the call, at the CS1000E SIP Gateway, use pcap tool (ethereal/wireshark) to make sure that all SIP request/response messages are received properly.
- Verify Codec and SIP trunk status when call is established under CS1000E call server by tracing DID number

LD 80
.trac 0 496856

ACTIVE VTN 096 0 01 06
ORIG VTN 096 0 01 06 KEY 0 SCR MARP CUST 0 DN 496856 TYPE 2002P2
SIGNALING ENCRYPTION: INSEC
MEDIA ENDPOINT IP: 47.248.101.120 PORT: 5200
TERM VTN 100 0 00 31 VTRK IPTI RMBR 100 32 OUTGOING VOIP GW CALL
FAR-END SIP SIGNALING IP: 217.110.230.98
FAR-END MEDIA ENDPOINT IP: 217.110.230.97 PORT: 6478
FAR-END VendorID: Not available
MEDIA PROFILE: **CODEC G.711 A-LAW** PAYLOAD 20 ms VAD OFF
RFC2833: RXPT 101 TXPT 101 DIAL DN 916139675258
MAIN_PM ESTD
TALKSLOT ORIG 21 TERM 53
QUEU NONE
CALL ID 511 941

---- ISDN ISL CALL (TERM) ----

CALL REF # = 416

BEARER CAP = VOICE

HLC =

CALL STATE = 10 ACTIVE

CALLING NO = 442033496856 NUM_PLAN:E164 TON:INTERNATIONAL

ESN:UNKNOWN

CALLED NO = 16139675258 NUM_PLAN:E164 TON:INTERNATIONAL

ESN:UNKNOWN

- Verify SIP Trunk is released when DID number is released the call by tracing that DID number under CS1000E call server

LD 80

.trac 0 496856 (DID number)

IDLE VTN 096 0 01 06 MARP

Conclusion

All of the executed test cases have passed and met the objectives outlined in **Section 6.1**, with some exceptions outlined in **Section 6.2**. The outstanding issues are being investigated by Frontier and Avaya design teams. Some of these issues are considered as exceptions. The Frontier Communication System is considered compliant with Communication Server 1000E release 6.0.

Additional References

Product documentation for Avaya products may be found at:

<http://support.nortel.com/go/main.jsp>

- [1] *Communication Server 1000E Overview Release 6.0, Revision 03.04, October 2009, Document Number NN43041-110*
- [2] *Product Compatibility Matrix release 5.0/5.5/6.0, Revision 01.07, February 2010, Document Number NN43001-140*
- [3] *Communication Server 1000 Network Routing Service Fundamentals, Release 6.0, Revision 01.04, Jun 2009, Document Number NN43001-130*
- [4] *Communication Server 1000 Unified Communications Management Common Services Fundamentals, Revision 03.05, February 2010, Document Number NN43001-116*
- [5] *Communication Server 1000 SIP Line Fundamentals, Release 6.0, Revision 01.08, February 10, Document Number NN43001-508*
- [6] *Communication Server 1000 Dialing Plans Reference, Release 6.0, Revision 03.09, June 2009, Document Number NN43001-283*

Appendixes

Appendix A: CS1000E CPPM Call Server RIs 6.00R Patches Installed

Id 143

CCBR000

.mdp issp

VERSION 4121

RELEASE 6

ISSUE 00 R +

DepList 1: core Issue: 02 (created: 2010-02-02 13:33:25 (est)) ALTERED

IN-SERVICE PEPS

PAT#	CR #	PATCH REF #	NAME	DATE	FILENAME	SPECINS
000	Q01976701-01	ISS1:1OF1	p28211_1	29/03/2010	p28211_1.cpl	NO
001	Q02029209	ISS1:1OF1	p28469_1	29/03/2010	p28469_1.cpl	NO
002	Q02023636	ISS1:1OF1	p28475_1	29/03/2010	p28475_1.cpl	NO
003	Q02041702	ISS1:1OF1	p28698_1	29/03/2010	p28698_1.cpl	NO
004	Q02027777	ISS1:1OF1	p28471_1	29/03/2010	p28471_1.cpl	NO
005	Q02038440	ISS1:1OF1	p28674_1	29/03/2010	p28674_1.cpl	NO
006	Q02034835	ISS1:1OF1	p28569_1	29/03/2010	p28569_1.cpl	YES

007	Q02040015	ISS1:1OF1	p28657_1	29/03/2010	p28657_1.cpl	NO
008	Q02094012	ISS1:1OF1	p29370_1	29/03/2010	p29370_1.cpl	YES
009	Q02039217-01	ISS1:1OF1	p28760_1	29/03/2010	p28760_1.cpl	NO
010	Q02031118	ISS1:1OF1	p28680_1	29/03/2010	p28680_1.cpl	NO
011	Q02096711	ISS1:1OF1	p29394_1	29/03/2010	p29394_1.cpl	NO
012	Q02024135-04	ISS1:1OF1	p28381_1	29/03/2010	p28381_1.cpl	NO
013	Q02021470-02	ISS1:1OF1	p28776_1	29/03/2010	p28776_1.cpl	NO
014	Q02043231	ISS1:1OF1	p28712_1	29/03/2010	p28712_1.cpl	NO
015	Q02083027	ISS1:1OF1	p29233_1	29/03/2010	p29233_1.cpl	NO
016	Q02033321	ISS1:1OF1	p28801_1	29/03/2010	p28801_1.cpl	NO
017	Q02033951	ISS1:1OF1	p28579_1	29/03/2010	p28579_1.cpl	NO
018	Q01782930-01	ISS1:1OF1	p24964_1	29/03/2010	p24964_1.cpl	NO
019	Q02028560-04	ISS1:1OF1	p28564_1	29/03/2010	p28564_1.cpl	NO
020	Q02041981	p28695_1	p28719_1	29/03/2010	p28719_1.cpl	NO
021	Q02033139	ISS1:1OF1	p28582_1	29/03/2010	p28582_1.cpl	NO
022	Q02039181	ISS1:1OF1	p28644_1	29/03/2010	p28644_1.cpl	NO
023	Q02030977	ISS1:1OF1	p28507_1	29/03/2010	p28507_1.cpl	NO
024	Q02076740	ISS1:1OF1	p29154_1	29/03/2010	p29154_1.cpl	NO
025	Q02012100-06	ISS1:1OF1	p29368_1	29/03/2010	p29368_1.cpl	NO
026	Q02035555	ISS1:1OF1	p28814_1	29/03/2010	p28814_1.cpl	NO
027	Q02021384-01	ISS1:1OF1	p28615_1	29/03/2010	p28615_1.cpl	NO
028	Q02032955-02	ISS1:1OF1	p28529_1	29/03/2010	p28529_1.cpl	NO
029	Q02055997	ISS1:1OF1	p28895_1	29/03/2010	p28895_1.cpl	NO
030	Q02022264	ISS1:1OF1	p28486_1	29/03/2010	p28486_1.cpl	NO
031	Q02031323-01	ISS1:1of1	p28546_1	29/03/2010	p28546_1.cpl	NO
032	Q01987279-02	ISS1:1OF1	p28416_1	29/03/2010	p28416_1.cpl	NO
033	Q00349046-03	ISS1:1OF1	p17588_1	29/03/2010	p17588_1.cpl	NO
034	Q02049121-01	ISS1:1OF1	p28819_1	29/03/2010	p28819_1.cpl	NO
035	Q02029228-01	ISS1:1OF1	p28681_1	29/03/2010	p28681_1.cpl	YES
036	Q02071626	ISS1:1OF1	p29163_1	29/03/2010	p29163_1.cpl	NO
037	Q02020526	ISS1:1OF1	p28537_1	29/03/2010	p28537_1.cpl	NO
038	Q02034783-01	p28596	p28594_1	29/03/2010	p28594_1.cpl	YES
039	Q02039994	ISS1:1OF1	p28690_1	29/03/2010	p28690_1.cpl	NO
040	Q01986974-05	ISS1:1OF1	p28821_1	29/03/2010	p28821_1.cpl	YES
041	Q02035396	ISS1:1OF1	p28675_1	29/03/2010	p28675_1.cpl	NO
042	Q02035822-01	ISS1:1OF1	p29212_1	29/03/2010	p29212_1.cpl	NO

043	Q02097631	ISS1:1OF1	p28328_1	29/03/2010	p28328_1.cpl	NO
044	Q02093188	ISS1:1OF1	p29352_1	29/03/2010	p29352_1.cpl	NO
045	Q02071451	ISS1:1OF1	p29164_1	29/03/2010	p29164_1.cpl	NO
046	Q01983521-04	ISS1:1OF1	p27616_1	29/03/2010	p27616_1.cpl	NO
047	Q02077909	ISS1:1of1	p29272_1	29/03/2010	p29272_1.cpl	NO
048	Q02073690	ISS1:1OF1	p29208_1	29/03/2010	p29208_1.cpl	NO
049	Q02092594	ISS1:1OF1	p27830_1	29/03/2010	p27830_1.cpl	NO
050	Q02031359	p28679	p28725_1	29/03/2010	p28725_1.cpl	YES
051	Q02079849	ISS1:1OF1	p29238_1	29/03/2010	p29238_1.cpl	NO
052	Q02031959	ISS1:1OF1	p28728_1	29/03/2010	p28728_1.cpl	NO
053	Q02038675	ISS1:1OF1	p28665_1	29/03/2010	p28665_1.cpl	YES
054	Q02100914	ISS1:1OF1	p28597_1	29/03/2010	p28597_1.cpl	NO
055	Q02024455-01	ISS1:1OF1	p28717_1	29/03/2010	p28717_1.cpl	NO
056	Q02020734-02	ISS1:1OF1	p28668_1	29/03/2010	p28668_1.cpl	NO
057	Q02044341	ISS1:1OF1	p28957_1	29/03/2010	p28957_1.cpl	NO
058	Q02064503	ISS1:1OF1	p29196_1	29/03/2010	p29196_1.cpl	NO
059	Q02095838	1SS1:1OF1	p28852_1	29/03/2010	p28852_1.cpl	NO
060	Q02043398	ISS1:1OF1	p28869_1	29/03/2010	p28869_1.cpl	NO
061	Q02033000	ISS1:1of1	p28736_1	29/03/2010	p28736_1.cpl	NO
062	Q02089407	ISS1:1OF1	p29311_1	29/03/2010	p29311_1.cpl	NO
063	Q01981776-01	ISS1:1OF1	p29065_1	29/03/2010	p29065_1.cpl	NO
064	Q02017013-01	ISS1:1OF1	p28313_1	29/03/2010	p28313_1.cpl	NO
065	Q02077171	ISS1:1OF1	p29169_1	29/03/2010	p29169_1.cpl	NO
066	Q02031502	ISS1:1OF1	p28832_1	29/03/2010	p28832_1.cpl	YES
067	Q02086333	ISS1:1OF1	p29262_1	29/03/2010	p29262_1.cpl	YES
068	Q02102219-01	ISS1:1OF1	p29464_1	29/03/2010	p29464_1.cpl	NO
069	Q02092223	ISS1:1OF1	p29343_1	29/03/2010	p29343_1.cpl	NO
070	Q02077848-01	ISS1:1OF1	p29320_1	29/03/2010	p29320_1.cpl	NO

Appendix B: CS1000E CPPM Signaling Server RIs

6.00.18 Patches Installed

Product Release: 6.00.18.00

In system patches: 7

PATCH#	NAME	IN_SERVICE	DATE	SPECINS	TYPE	RPM
--------	------	------------	------	---------	------	-----

6	p28774_1	Yes	17/03/10	NO	FRU	nortel-cs1000-Jboss-Quantum-6.00.18.00-00.i386
---	----------	-----	----------	----	-----	--

7	p28797_1	Yes	17/03/10	NO	FRU	nortel-cs1000-Jboss-Quantum-6.00.18.00-00.i386
20	p22968_1	Yes	24/03/10	NO	FRU	nortel-cs1000-pi-control-1.00.00.00-00.noarch
22	p28415_1	Yes	24/03/10	NO	FRU	nortel-cs1000-pi-control-1.00.00.00-00.noarch
23	p25946_1	Yes	24/03/10	NO	FRU	nortel-cs1000-pi-control-1.00.00.00-00.noarch
24	p25529_1	Yes	26/03/10	NO	FRU	nortel-cs1000-pi-control-1.00.00.00-00.noarch
25	p27408_1	Yes	26/03/10	NO	FRU	nortel-cs1000-pi-control-1.00.00.00-00.noarch

In System service updates: 18

PATCH#	IN_SERVICE	DATE	SPECINS	REMOVABLE	NAME
0	Yes	17/03/10	YES	YES	nortel-cs1000-linuxbase-6.00.18.63-02.i386.000
2	Yes	17/03/10	NO	YES	nortel-cs1000-dmWeb-6.00.18.62-00.i386.001
3	Yes	17/03/10	NO	YES	nortel-cs1000-auth-6.00.18.62-00.i386.000
4	Yes	17/03/10	NO	YES	nortel-cs1000-ISECSH-6.00.18.62-00.i386.000
5	Yes	17/03/10	YES	YES	nortel-cs1000-patchWeb-6.00.18.63-01.i386.000
8	Yes	17/03/10	NO	YES	nortel-cs1000-gk-6.00.18.63-00.i386.000
9	Yes	17/03/10	NO	YES	nortel-cs1000-sps-6.00.18.63-00.i386.000
10	Yes	17/03/10	NO	YES	nortel-cs1000-tps-6.00.18.63-00.i386.000
11	No	17/03/10	NO	YES	nortel-cs1000-bcc_6-0-6.00.18.63-01.i386.000
12	No	17/03/10	NO	YES	nortel-cs1000-cs1000WebService_6-0-6.00.18.63-01.i386.000
13	Yes	17/03/10	NO	YES	nortel-cs1000-shared-general-6.00.18.62-00.i386.000
14	Yes	17/03/10	NO	YES	nortel-cs1000-shared-pbx-6.00.18.62-00.i386.000
15	No	17/03/10	NO	YES	nortel-cs1000-emWeb_6-0-06.00.18.63-01.i386.001
16	Yes	17/03/10	NO	YES	nortel-cs1000-pd-6.00.18.62-00.i386.000
17	Yes	16/04/10	NO	YES	nortel-cs1000-vtrk-6.00.18.65-017.i386.000
18	Yes	17/03/10	NO	YES	nortel-cs1000-csv-6.00.18.62-00.i386.000
19	No	17/03/10	NO	YES	nortel-cs1000-csmWeb-6.00.18.62-00.i386.001
21	Yes	17/03/10	NO	YES	nortel-cs1000-dbcom-6.00.18.65-01.i386.001
30	Yes	26/03/10	NO	YES	nortel-cs1000-vtrk-6.00.18.065-016.i386.001

Appendix C: Configure SIP trunk in CS1000 using overlays

Procedure summary

This information is provided as a simple summary of tasks to complete when configuring IP Peer Networking, but it does not replace the full details provided in the IP Peer Networking Guide.

No.	Overlay	Element Management	Action
1	LD 97		Define a virtual super loop
2	LD 17	Select Configuration/D-Channel link	Create a virtual D-channel
3	LD 15	Select Configuration/Customer Explorer link	Define the customer to support ISDN
4	LD 16	Select Configuration/Customer Explorer /Add Route	Create a virtual service route
5	LD 14	Select Configuration/Customer Explorer /Add Trunk	Create virtual trunks

Define a virtual superloop

Use Overlay 97

Prompt	Response	Description
REQ	CHG	
TYPE	SUPL	Configuration data block
SUPL	V100	Virtual superloop number (96 - 112 and multiple of 4 for 11C systems.)//CS 1000E not vloop100

Create a virtual D-channel

Use Overlay 17

Prompt	Response	Description
REQ	CHG	
TYPE	ADAN	Configuration data block
ADAN	NEW DCH 100	Add a primary D-Channel port 100
CTYP	DCIP	D-channel is over IP
DES	VIRTUAL_TRK	Description
USR	ISLD	Integrated services signaling link dedicated
IFC	SL1	Interface type is Meridian 1 – Meridian 1
ISLM	4000	Integrated services signaling link maximum
SIDE	USR	Slave to the controller (USR).
RLS	25	X11 software release of far-end.//not need

RCAP	ND2	Name display format 2//not need
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Define a customer with ISDN support

Use Overlay 15

Prompt	Response	Description
REQ	NEW	
TYPE	CDB	Customer data block
CUST	0	Customer number
ANAT	1111	ANI Attendant billing number for making ANI calls
ANLD	111	ANI listed directory number
ISDN	YES	Customer is equipped with ISDN.
VPNI	1	Virtual private network identifier//important
PNI	1	Private network identifier.//important

Define a virtual service route

Use Overlay 16

Prompt	Response	Description
REQ	NEW	
TYPE	RDB	Route data block
CUST	0	Customer number
ROUT	100	Route number
DES	VTRK	Designator field for trunk
TKTP	TIE	TIE trunk only, allowed between SL-1
ICOG	IAO	Incoming and outgoing
VTRK	YES	Virtual trunk route
ZONE	0	Zone for codec selection and bandwidth management
NODE	2000	Node ID of signaling server of this route.
PCID	SIP	Protocol ID for this route
ISDN	YES	ISDN option

MODE	ISLD	Route uses ISDN signaling link
DCH	100	D-channel number for this route
PNI	1	Customer private network identifier.
IFC	SL 1	Interface type : Meridian 1 to Meridian 1
NCNA	YES	Network calling name allowed.
NCRD	YES	Network call redirection.
CHTY	BCH	B-channel type.
CTYP	CDP	Coordinated dialing plan

Define virtual trunks

Use Overlay 14

Prompt	Response	Description
REQ	NEW 32	
TYPE	IPTI	IP trunk
TN	100 0 0 0	Virtual card and channel number
DES	VTRK	Designator field for trunk
CUST	0	Customer number
RTMB	100 1	Route number and member number.
STRI	IMM	Start arrangement incoming
STRO	IMM	Start arrangement outgoing
TGAR	1	Trunk group access restriction.
CHID	1	Channel ID for trunk

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