



**Application Notes for configuring Logically IVR from  
Maximum Network Solutions with Avaya Communication  
Server 1000E R7.6 and Avaya Aura® Session Manager R6.3  
- Issue 1.0**

**Abstract**

These Application Notes describe the configuration steps necessary for provisioning Maximum Network Solutions Logically IVR to successfully interoperate with Avaya Communication Server 1000E R7.6 and Avaya Aura® Session Manager R6.3.

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

These Application Notes describe a compliance-tested configuration of the interoperability of Logically IVR from Maximum Network Solutions to successfully interoperate with Avaya Communication Server 1000E R7.6 and Avaya Aura® Session Manager R6.3. Maximum Network Solutions Logically IVR is a voice processing platform capable of supporting both DTMF (touch tone) and natural speech. Maximum Network Solutions Logically IVR provides a platform on which voice applications and speech self-service applications can be installed enabling the automation of telephone transactions and is available as a pre-configured, out of the box solution.

## 1. General Test Approach and Test Results

The Maximum Network Solutions Logically IVR solution (LIVR) sits on Solaris R10 or Linux R6.3 platform installed on a Sun Microsystems server and is administered over a telnet session using a program such as PuTTY or Reflections. Each solution is pre-configured according to the end-users specifications, the configuration regarding connectivity to the Avaya Communication Server 1000E (CS1000E) is contained in a configurable ini file called Control.ini, an example of which can be seen in the **Appendix** of these Application Notes. The test approach was to successfully place calls to the Logically box over SIP trunks between the Logically and the CS1000E. Calls from the CS1000E to the Logically route through the Session Manager.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly-defined test plan focuses on exercising APIs and/or standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute full product performance or feature testing performed by DevConnect members, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a DevConnect member's solution.

### 1.1. Interoperability Compliance Testing

During interoperability compliance testing the following types of calls were made and features of Logically covered.

- Inbound calls to Logically IVR from CS1000E 1140E IP and SIP deskphones.
- Inbound calls to Logically IVR front-ended by Avaya Callpilot.
- Inbound calls to Logically IVR using menu to get transferred.
- Calls placed into Logically from PSTN users.
- Calls transferred to Logically from CS1000E 1140E IP and SIP deskphones.
- Calls transferred to Logically from PSTN users.
- Calls transferred to Logically from an Avaya Callpilot menu.
- Outbound calls to CS1000E 1140E IP and SIP deskphones from Logically.
- Automatic Speech Recognition (ASR) test on Logically.
- Text to Speech (TTS) test on Logically.

## 1.2. Test Results

All tests outlined in the Test Plan document passed successfully. Below are listed observations following the compliance test of this solution.

1. No trunks were terminated directly on the Logically IVR box. All calls to Logically IVR came through the CS1000E. External trunk calls over QSig or SIP terminated on the CS1000E and were then routed across to the Logically IVR via Session Manager. Note that for the compliance testing these PSTN trunks were simulated by connecting to a second CS1000E PBX.
2. When a caller dials into the Logically IVR and gets transferred out using “outbound”, “blind” or “transfer”, the CLID being passed to the caller from the Logically IVR is dynamically set by the Logically IVR box and in this compliance testing this was set to “2500”.
3. When the transfer was complete or the call was answered from the blind or outbound calls, the CLID info was not updated to the original callers ID, this remained as “2500” and this is designed by Logically to behave this way.
4. The original caller CLID does get updated with the transferred parties’ number. This was an intermittent issue – on most instances there was no issue with CLID updating on blind transfer. This was for both the QSIG and SIP trunk callers.
5. The CLID can be manipulated by changes on the D-Channels. All CLID behaves as expected for internal calls so issues with external CLID may be carrier related or in this case the way in which the QSIG-SIP trunks are setup to deal with CLID/Names.

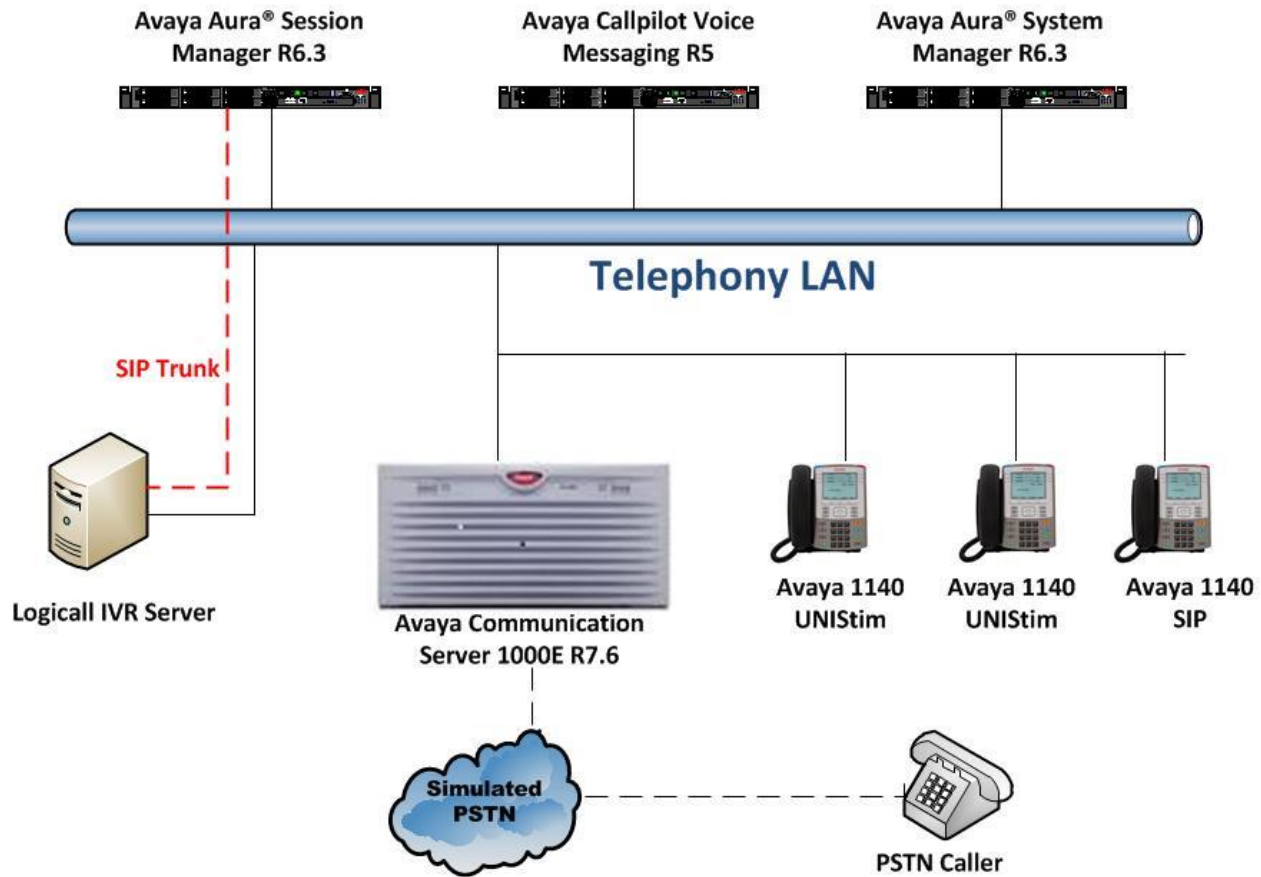
## 1.3. Support

For more information on Maximum Network Solutions (Maxnet) and product support visit <http://www.maxnet.co.uk>. The following is the contact information for Maxnet:

Maximum Network Solutions  
The Old Granary, The Square, Sheffield,  
South Yorkshire, S26 5QN  
+44 1909 774477  
[www.maxnet.co.uk](http://www.maxnet.co.uk)

## 2. Reference Configuration

The configuration in **Figure 1** was used to compliance test Logically IVR with Avaya Communication Server 1000E using Session Manager and SIP trunks to pass calls from the CS1000E to the Logically IVR.



**Figure 1: Connection of Maximum Network Solutions Logically IVR with Avaya Aura® Session Manager and Avaya Communication Server 1000E R7.6.**

### 3. Equipment and Software Validated

The following equipment and software were used for the sample configuration provided:

Equipment/Software	Release/Version
Avaya Communication Server 1000E on CPPM	R7.6 SP2 (See Appendix for list of patches)
Avaya Aura® System Manager running on Avaya Virtual Server	System Manager 6.3.0 - FP2 Build No. - 6.3.0.8.5682-6.3.8.1814 Software Update Revision No: 6.3.3.5.1719
Avaya Aura® Session Manager running on Avaya Virtual Server	Session Manager R6.3 SM 6.3.3.0.633004
Avaya 1140 UNISTim Deskphone	UNISTim V0625C8D
Avaya 1140 SIP Deskphone	SIP 04.03.12
Avaya 3904 Digital Deskphone	Core V2.4 Flash V9.4
Maximum Networks Logically IVR Server	LIVR V1.1

## 4. Configuration of Avaya Communication Server 1000E

The configuration operations illustrated in this section were performed using terminal access to the CS1000E using PuTTY. It is assumed a fully working CS1000E is in place with the necessary licensing. For all other provisioning information, such as Administering Avaya CS1000E, refer to product documentation in **Section 11** of these Application Notes.

**Note:** The configuration of PSTN trunks and routes are outside the scope of these Application Notes.

**Note:** Not all prompts need an answer. The prompts outlined below are mandatory for a basic configuration. Accept the default responses for all other prompts by pressing the Return key.

**Note:** A full printout of the SIP D-Channel, Route and Trunk information used for the compliance testing is included in the **Appendix A** of these Application Notes.

### 4.1. Verify Licences

To ensure the CS1000E is licensed for SIP, use **LD 22** and type **SLT** at the **REQ** prompt. Check for **SIP ACCESS PORTS** (in bold below).

Prompt	Response	Description
>	<b>LD 22</b>	Enter Overlay 22
REQ	<b>SLT</b>	
System type is - Communication Server 1000E/CPPM Linux		
CPPM - Pentium M 1.4 GHz		
IPMGs Registered: 1		
IPMGs Unregistered: 0		
IPMGs Configured/unregistered: 0		
TRADITIONAL TELEPHONES	2000	LEFT 1992 USED 8
DECT USERS	2000	LEFT 2000 USED 0
IP USERS	4000	LEFT 3978 USED 22
BASIC IP USERS	2000	LEFT 1998 USED 2
TEMPORARY IP USERS	2000	LEFT 2000 USED 0
DECT VISITOR USER	2000	LEFT 2000 USED 0
ACD AGENTS	2000	LEFT 1995 USED 5
MOBILE EXTENSIONS	2000	LEFT 2000 USED 0
TELEPHONY SERVICES	2000	LEFT 2000 USED 0
CONVERGED MOBILE USERS	2000	LEFT 2000 USED 0
AVAYA SIP LINES	2000	LEFT 1997 USED 3
THIRD PARTY SIP LINES	2000	LEFT 1998 USED 2
PCA	2000	LEFT 2000 USED 0
ITG ISDN TRUNKS	2000	LEFT 2000 USED 0
H.323 ACCESS PORTS	2000	LEFT 1990 USED 10
AST	2000	LEFT 1981 USED 19
SIP CONVERGED DESKTOPS	2000	LEFT 2000 USED 0
SIP CTI TR87	2000	LEFT 1992 USED 8
<b>SIP ACCESS PORTS</b>	<b>2000</b>	<b>LEFT 1970 USED 30</b>
RAN CON	2000	LEFT 2000 USED 0

MUS CON	2000	LEFT	2000	USED	0
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## 4.2. Configuring a SIP Connection on CS1000E

To configure the SIP connection there are a number of steps.

- Create a D-channel for the SIP trunk
- Create Route Data Block
- Add TIE Trunks

### 4.2.1. Create a D-Channel

Use the **CHG** command in **LD 17** to create a D-channel for the SIP connection. In the example below, D-Channel 66 (i.e. **DCH 66**) was created. At the **CTYP** prompt, enter **DCIP**. This signifies the SIP D-Channel.

#### LD 17

Prompt	Response	Description
>	<b>LD 17</b>	Enter Overlay 17
REQ	<b>CHG</b>	Change
TYPE	ADAN	Change the Action Device and Number
ADAN	NEW	Create New Action Device and Number
TYPE	<b>DCH 66</b>	Create new D-Channel 66
<b>CTYP</b>	<b>DCIP</b>	Card type is IP D-Channel
USR	ISDL	Integrated Services Digital Line
IFC	SL1	D-Channel interface type

### 4.2.2. Create Route Data Block

Use the **NEW** command in **LD 16** to create a Route Data Block. The route created is a **TIE** route in order to connect to the Logically IVR. Ensure **VTRK** is set to **YES** and **PCID** is **SIP**. Ensure that the other values highlighted are configured correctly. A complete printout of all prompts can be found in **Appendix A** of these Application Notes.

#### LD 16

Prompt	Response	Description
>	<b>LD 16</b>	Enter Overlay 16
REQ	<b>NEW</b>	Create new
TYPE	RDB	Route Data block
CUST	0	Customer Number as defined in LD15
<b>ROUT</b>	<b>22</b>	Route Number
<b>TKTP</b>	<b>TIE</b>	Route Type
<b>VTRK</b>	<b>YES</b>	Virtual Route
<b>PCID</b>	<b>SIP</b>	Protocol ID for route
<b>NODE</b>	<b>111</b>	Node setup during the installation of the CS1000E
DTRK	NO	Digital Trunk Route
<b>ISDN</b>	<b>YES</b>	Integrated Services Digital Network
<b>MODE</b>	<b>ISDL</b>	mode of operation
<b>IFC</b>	<b>SL1</b>	Interface type
ACOD	8022	Access Code for trunk route



### 4.2.3. Adding TIE Trunks

Use the **NEW** command in **LD 14** to add (**IPTI**) **TIE** trunks to the new route created in **Section 5.2.2**. If adding multiple trunks for each route, use **NEW XX**, where **XX** is the number of trunks. In the example below **10** trunks were added.

#### LD 14

Prompt	Response	Description
>	<b>LD 14</b>	Enter Overlay 14
REQ	<b>NEW 10</b>	Create 10 New Trunks
TYPE	<b>IPTI</b>	IP TIE trunk
TN	100 0 3 0	Loop Shelf Card Unit
CUST	0	Customer Number as defined in LD15
RTMB	22 1	Route number and Member number

## 4.3. Configure a Coordinated Dialing Plan

In order to setup a Coordinated Dialing Plan (CDP), both a route list index and a CDP are added.

### 4.3.1. Create a Route List Index

Use the **NEW** command in **LD 86** to create a **RLI**. Enter the route (**ROUT**) that was created in **Section 5.2.2**.

#### LD 86

Prompt	Response	Description
> <b>LD 86</b>	Enter Overlay 86	
REQ	<b>NEW</b>	Create New
CUST	0	Customer Number as defined in LD15
FEAT	<b>RLB</b>	Route list Block
TYPE	<b>RLI</b>	Route list Index
RLI	22	Route list Index number
ENTR	0	First entry for the RLI
ROUT	22	Enter the route number

### 4.3.2. Create CDP

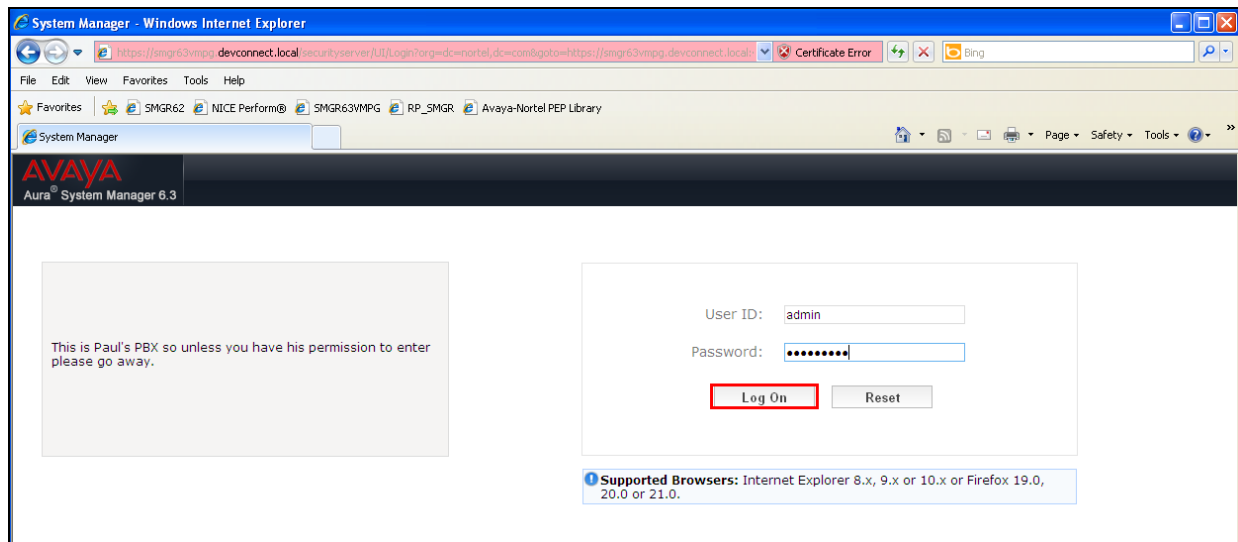
Use the **NEW** command in **LD 87** to create a **CDP** entry for the Trio Enterprise. For each extension, a CDP entry needs to be created. In the example below, the **DSC** is **6000**, **FLEN** is **4** and the **RLI** is **22**. The RLI number used is the one created in **Section 5.3.1**.

#### LD 87

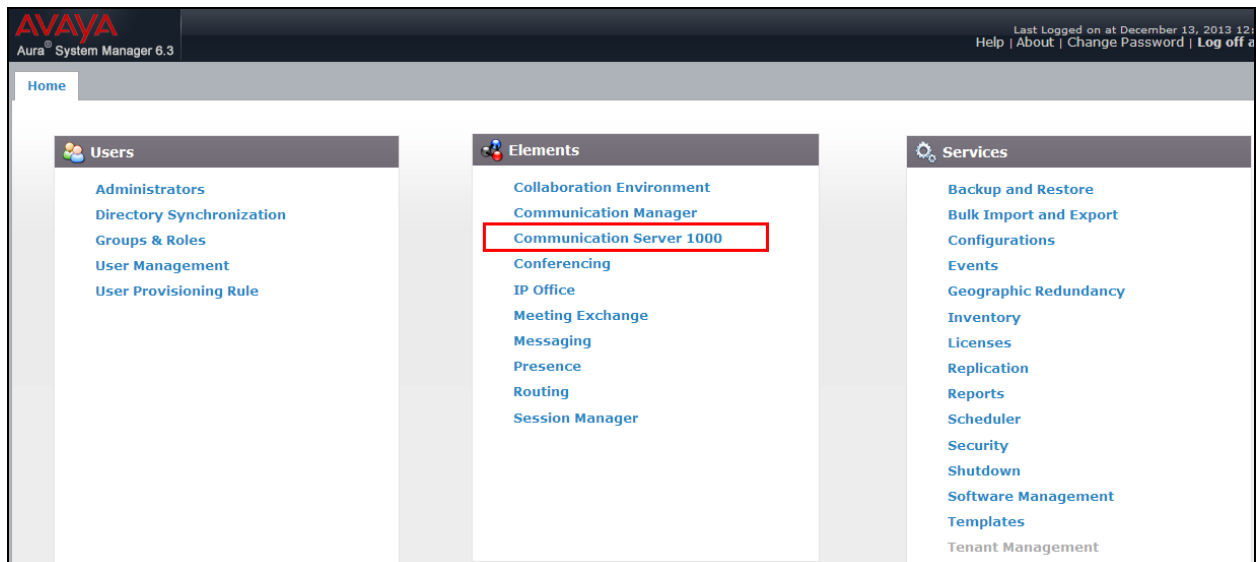
Prompt	Response	Description
>	LD 87	Enter Overlay 87
REQ	<b>NEW</b>	Create new
CUST	0	Customer Number as defined in LD15
FEAT	<b>CDP</b>	Coordinated dialing plan
TYPE	<b>DSC</b>	Distance Steering code
<b>DSC</b>	<b>6000</b>	Distant Steering code
<b>FLEN</b>	<b>4</b>	Flexible Length number of digits
RLI	22	Route list index Number

## 5. Configure Avaya Communication Server 1000E Signalling Server

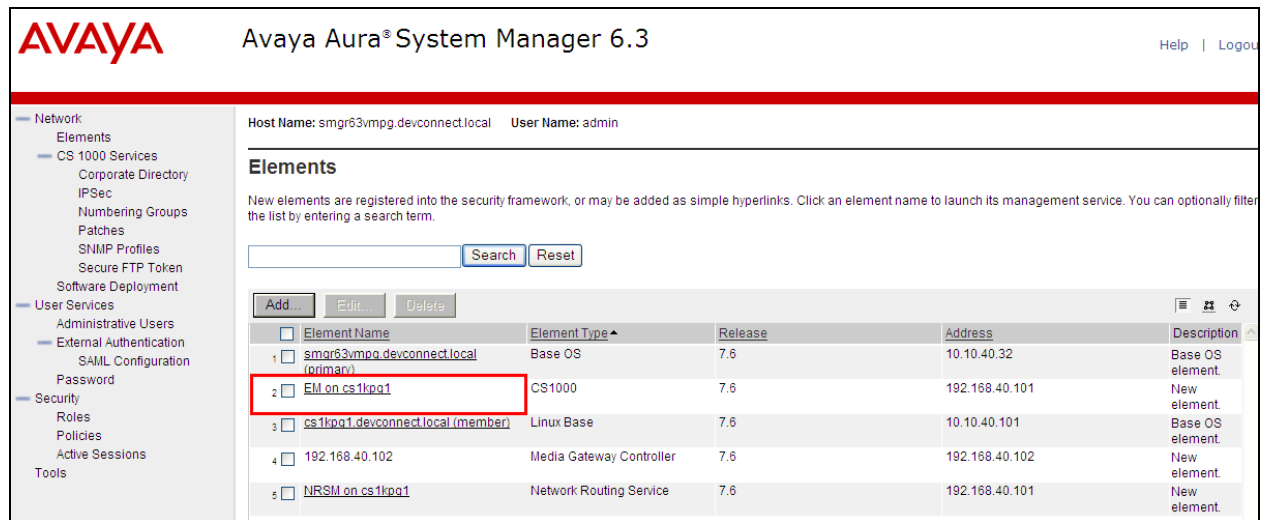
Access to the CS1000E Signalling Server is achieved by logging into System Manager using a Web Browser by entering **http://<FQDN>/SMGR**, where <FQDN> is the fully qualified domain name of System Manager or **http://<IP Address>/SMGR**. Log in using appropriate credentials.



Once logged in click on **Communication Server 1000** as highlighted



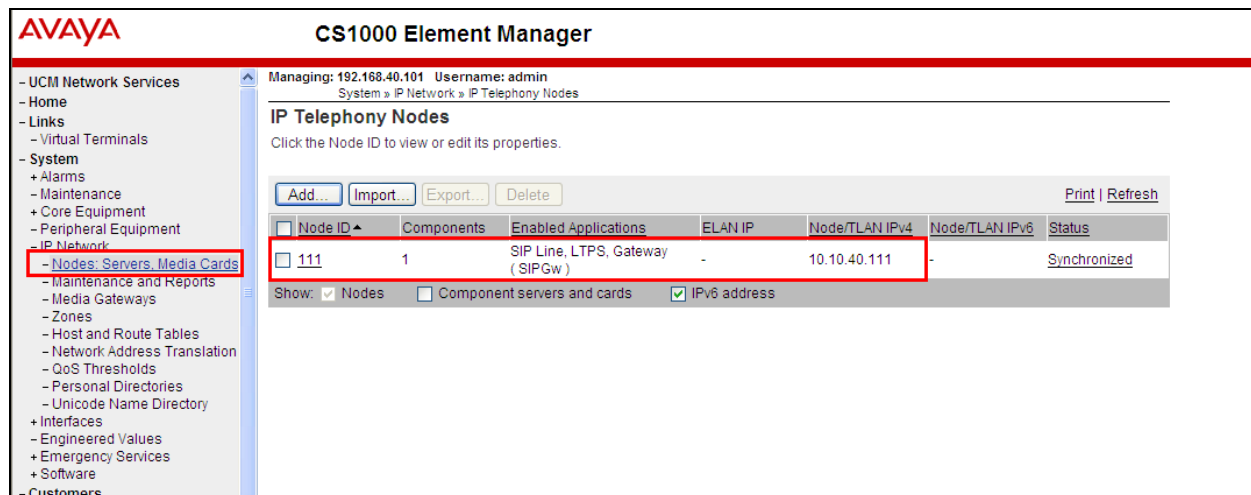
The following screen appears showing the various **Elements**, select **EM on cs1kpg** (note this name may appear different depending on the system).



The screenshot shows the Avaya Aura System Manager 6.3 interface. The left sidebar contains a navigation tree with categories like Network, User Services, Security, and Tools. The main content area is titled 'Elements' and displays a table of system elements. A red box highlights the second row of the table, which is 'EM on cs1kpg1'.

	Element Name	Element Type	Release	Address	Description
1	smgr63vmpg.devconnect.local (primary)	Base OS	7.6	10.10.40.32	Base OS element.
2	EM on cs1kpg1	CS1000	7.6	192.168.40.101	New element.
3	cs1kpg1.devconnect.local (member)	Linux Base	7.6	10.10.40.101	Base OS element.
4	192.168.40.102	Media Gateway Controller	7.6	192.168.40.102	New element.
5	NRSM on cs1kpg1	Network Routing Service	7.6	192.168.40.101	New element.

Navigate to **IP Network** → **Nodes Servers and Media Cards** in the left window and select the Node associated with the CS1000E, in the example below this **Node ID** is **111**. Open this node by clicking on **111** highlighted below.



The screenshot shows the CS1000 Element Manager interface. The left sidebar has a navigation tree with categories like UCM Network Services, Home, Links, System, and Customers. The main content area is titled 'IP Telephony Nodes' and displays a table of nodes. A red box highlights the first row of the table, which is Node ID 111.

Node ID	Components	Enabled Applications	ELAN IP	Node/TLAN IPv4	Node/TLAN IPv6	Status
111	1	SIP Line, LTPS, Gateway (SIPGw)	-	10.10.40.111	-	Synchronized

Select **Gateway (SIPGw)** highlighted.

**AVAYA CS1000 Element Manager**

Managing: 192.168.40.101 Username: admin  
System » IP Network » IP Telephony Nodes » Node Details

**Node Details (ID: 111 - SIP Line, LTPS, Gateway (SIPGw))**

Subnet mask: 255.255.255.0 \* Subnet mask: 255.255.255.0 \*  
Node IPv6 address:

**IP Telephony Node Properties**

- Voice Gateway (VGW) and Codecs
- Quality of Service (QoS)
- LAN
- SNTP
- Numbering Zones
- MCDN Alternative Routing Treatment (MALT) Causes

**Applications (click to edit configuration)**

- SIP Line
- Terminal Proxy Server (TPS)
- Gateway (SIPGw)**
- Personal Directories (PD)
- Presence Publisher
- IP Media Services

\* Required Value. [Save] [Cancel]

**Associated Signaling Servers & Cards**

Select to add [Add] [Remove] [Make Leader] [Print] [Refresh]

Hostname	Type	Deployed Applications	ELAN IP	TLAN IPv4	Role
cs1kpg1	Signaling_Server	SIP Line, LTPS, Gateway (SIP/H323), PD, Presence Publisher, IP Media Services	192.168.40.101	10.10.40.101	Leader

Show: ☐ IPv6 address

Enter the correct **SIP domain name**, note this will be referenced again in **Section 7.2**.

**AVAYA CS1000 Element Manager**

Managing: 192.168.40.101 Username: admin  
System » IP Network » IP Telephony Nodes » Node Details » Virtual Trunk Gateway Configuration

**Node ID: 111 - 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:** devconnect.local \*

**Local SIP port:** 5060 \* (1 - 65535)

**Gateway endpoint name:** CS1KPG1 \*

**Gateway password:** \*

**Application node ID:** 111 \* (0-9999)

Enable failsafe NRS: ☐

Note: FailSafe NRS cannot be enabled, if all servers in the node have NRS application deployed.

**Virtual Trunk Network Health Monitor**

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

Monitor IP: [Add]

Monitor addresses: [Remove]

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

Scroll down to **Proxy Or Redirect Server: Proxy Server Route 1** and enter the IP Address of the Session Manager for the **Primary TLAN IP address**. Ensure the **Port** number is set to **5060** and the **Transport protocol** is set to **TCP**, note this will be referenced again in **Section 7.10**. Everything else can be left as default.

The screenshot displays the AVAYA CS1000 Element Manager web interface. The left sidebar contains a navigation tree with categories like UCM Network Services, System, Interfaces, Customers, Routes and Trunks, and Dialing and Numbering Plans. The main content area is titled 'Node ID: 111 - Virtual Trunk Gateway Configuration Details'. It features a breadcrumb trail: 'System » IP Network » IP Telephony Nodes » Node Details » Virtual Trunk Gateway Configuration'. Below this, there are tabs for 'General', 'SIP Gateway Settings', and 'SIP Gateway Services'. The 'SIP Gateway Settings' tab is active, showing the 'Proxy Or Redirect Server' configuration. This section includes a red-bordered box labeled 'Proxy Or Redirect Server:' containing a sub-section 'Proxy Server Route 1:'. Within this sub-section, the 'Primary TLAN IP address' is set to '10.10.40.34', the 'Port' is '5060', and the 'Transport protocol' is 'TCP'. There are also fields for 'Secondary TLAN IP address' (0.0.0.0) and its 'Port' (5060). A note at the bottom states: 'Note: Changes made on this page will NOT be transmitted until the Node is also saved.' There are 'Save' and 'Cancel' buttons at the bottom right.

Ensure the same details are filled in for the **Proxy Server Route 2**. Click on **Save** at the bottom right of the screen.

**AVAYA**

**CS1000 Element Manager**

- 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
  - Engineered Values
  - + Emergency Services
  - + Software
- Customers
- Routes and Trunks
  - Routes and Trunks
  - D-Channels
  - Digital Trunk Interface
- Dialing and Numbering Plans
  - Electronic Switched Network
  - Flexible Code Restriction

Managing: 192.168.40.101 Username: admin
System » IP Network » IP Telephony Nodes » Node Details » Virtual Trunk Gateway Configuration

**Node ID: 111 - Virtual Trunk Gateway Configuration Details**

General | SIP Gateway Settings | SIP Gateway Services

Options: ☐ Support registration  
☐ Tertiary CDS proxy

Proxy Server Route 2:

Primary TLAN IP address: 

The IP address can have either IPv4 or IPv6 format based on the value of "TLAN address type"

Port:  (1 - 65535)

Transport protocol:

Options: ☐ Registration not supported  
☒ Primary CDS proxy

CLID Presentation:

Country code (CCC):

Area code:  NPA in North America

\* Required Value.

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

Save

Cancel

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Logically\_SM63

Click on **Save** again as highlighted below.

**AVAYA CS1000 Element Manager**

Managing: 192.168.40.101 Username: admin  
System » IP Network » IP Telephony Nodes » Node Details

**Node Details (ID: 111 - SIP Line, LTPS, Gateway ( SIPGw ))**

Node ID: 111 \* (0-9999)  
Call server IP address: 192.168.40.101 \* TLAN address type: ☒ IPv4 only  
☐ IPv4 and IPv6

**Embedded LAN (ELAN)**  
Gateway IP address: 192.168.40.1 \*  
Subnet mask: 255.255.255.0 \*

**Telephony LAN (TLAN)**  
Node IPv4 address: 10.10.40.111 \*  
Subnet mask: 255.255.255.0 \*  
Node IPv6 address:

\* Required Value. **Save** Cancel

**Associated Signaling Servers & Cards**

Select to add Add Remove Make Leader Print Refresh

Hostname	Type	Deployed Applications	ELAN IP	TLAN IPv4	Role
<input type="checkbox"/> cs1kpg1	Signaling_Server	SIP Line, LTPS, Gateway (SIP/H323), PD, Presence	192.168.40.101	10.10.40.101	Leader

Select **Transfer Now** as shown below.

**AVAYA CS1000 Element Manager**

Managing: 192.168.40.101 Username: admin  
System » IP Network » IP Telephony Nodes » Node Saved

**Node Saved**

Node ID: 111 has been saved on the call server.  
The new configuration must also be transferred to associated servers and media cards.

**Transfer Now...** You will be given an option to select individual servers, or transfer to all.

Show Nodes You may initiate a transfer manually at a later time.

The following screen is displayed requiring that synchronization is performed followed by a restart of the Applications. Ensure the **Hostname** is ticked and click on **Start Sync**.

**AVAYA CS1000 Element Manager**

Managing: 192.168.40.101 Username: admin  
System » IP Network » IP Telephony Nodes » Synchronize Configuration Files

### Synchronize Configuration Files (Node ID <111>)

Note: Select components to synchronize their configuration files with call server data. This process transfers server INI files to selected components, and requires a restart\* of applications on affected server(s) when complete.

**Start Sync** Cancel Restart Applications [Print](#) | [Refresh](#)

<input checked="" type="checkbox"/>	Hostname	Type	Applications	Synchronization Status
<input checked="" type="checkbox"/>	cs1kpg1	Signalling_Server	SIP Line, LTPS, Gateway (SIP/H323), PD, Presence, Publisher, IP Media Services	Sync required

\* Application restart is only required for initial system configuration or if changes have been made to general LAN configurations, SNTP settings, SIP and H323 Gateway settings, network connectivity related parameters like ports and IP address, enabling or disabling services, or adding or removing application servers.

The following screen shows the **Sync in progress**.

**AVAYA CS1000 Element Manager**

Managing: 192.168.40.101 Username: admin  
System » IP Network » IP Telephony Nodes » Synchronize Configuration Files

### Synchronize Configuration Files (Node ID <111>)

Synchronization in progress. Status will be updated automatically.  
(You may also navigate away from this page and return to the [IP Telephony Nodes](#) list to verify completion.)

Start Sync Cancel [Print](#) | [Refresh](#)

	Hostname	Type	Applications	Synchronization Status
	cs1kpg1	Signalling_Server	SIP Line, LTPS, Gateway (SIP/H323), PD, Presence, Publisher, IP Media Services	Sync in progress

Once the Sync is completed select the **Hostname** again and click on **Restart Applications**. This will complete the Signalling Server configuration for Session Manager routing.



AVAYA

CS1000 Element Manager

Managing: 192.168.40.101 Username: admin

System » IP Network » IP Telephony Nodes » Synchronize Configuration Files

Synchronize Configuration Files (Node ID <111>)

Note: Select components to synchronize their configuration files with call server data. This process transfers server INI files to selected components, and requires a restart\* of applications on affected server(s) when complete.

Start Sync

Cancel

Restart Applications

Print | Refresh

<input checked="" type="checkbox"/>	Hostname	Type	Applications	Synchronization Status
<input checked="" type="checkbox"/>	cs1kpg1	Signaling_Server	SIP Line, LTPS, Gateway (SIP/H323), PD, Presence Publisher, IP Media Services	Sync required

\* Application restart is only required for initial system configuration or if changes have been made to general LAN configurations, SNTP settings, SIP and H323 Gateway settings, network connectivity related parameters like ports and IP address, enabling or disabling services, or adding or removing application servers.

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

Engineered Values

Emergency Services

Software

Customers

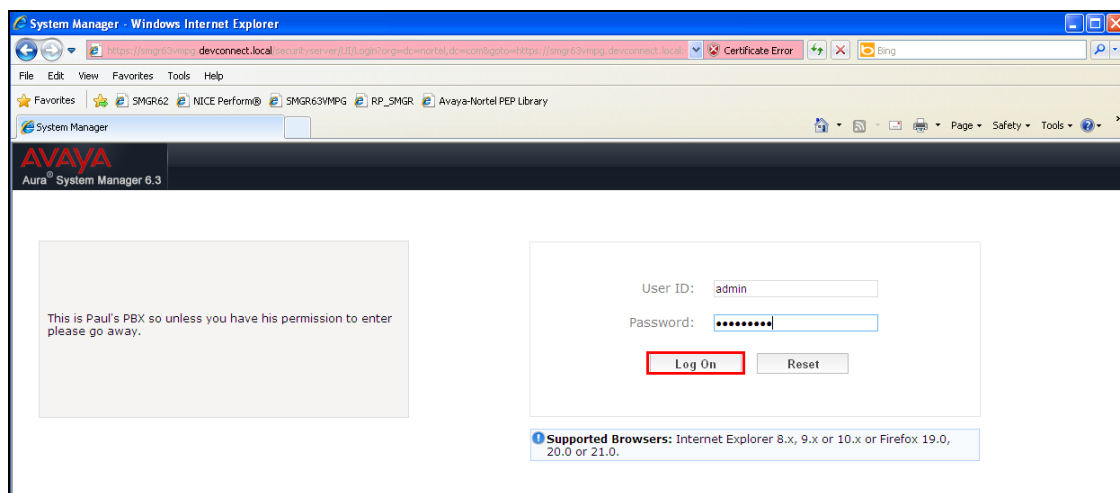
## 6. Configure Avaya Aura® Session Manager

This section provides the procedures for configuring Session Manager. Session Manager is configured via System Manager. The procedures include the following areas:

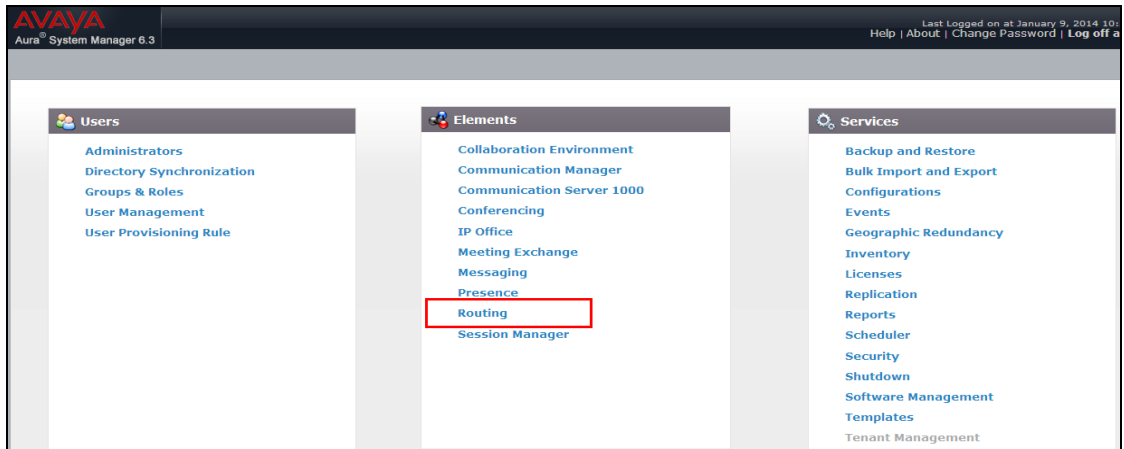
- Log in to Avaya Aura® Session Manager
- Administer SIP Domain
- Administer Location
- Configure Adaptation
- Administer SIP Entities
- Administer Routing Policies
- Administer Dial Patterns

### 6.1. Log in to Avaya Aura® System Manager

Access the System Manager using a Web Browser by entering **http://<FQDN>/SMGR**, where **<FQDN>** is the fully qualified domain name of System Manager or **http://<IP Address>/SMGR**. Log in using appropriate credentials.

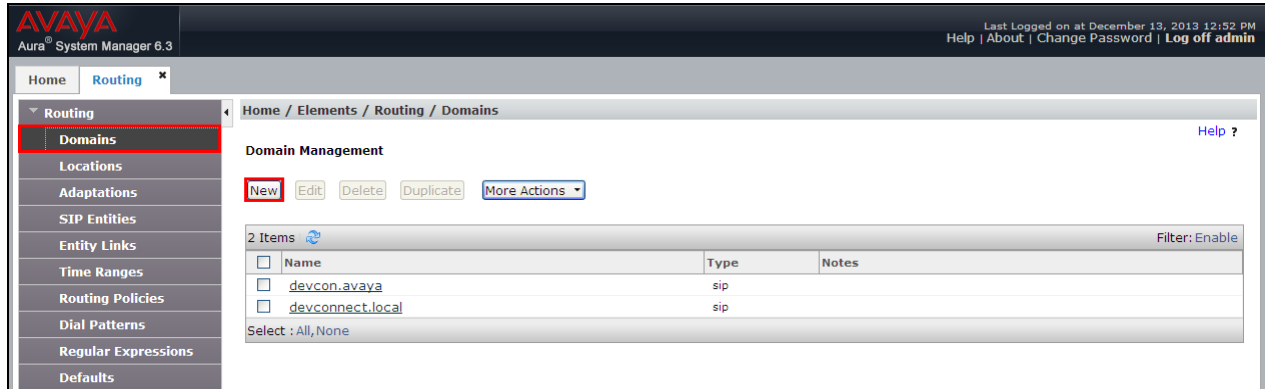


Once logged ion click on **Routing** as highlighted.

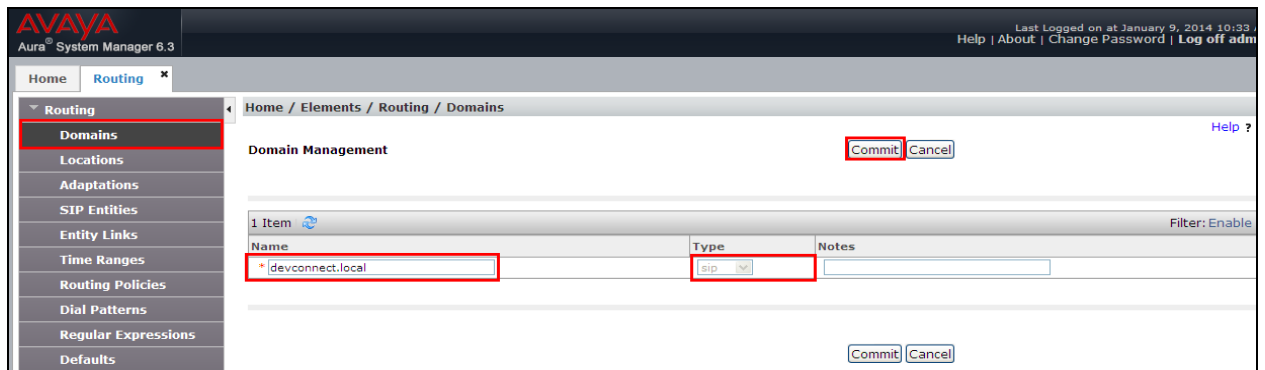


## 6.2. Administer SIP Domain

Click on **Domains** in the left window. If there is not a domain already configured click on **New** highlighted below.

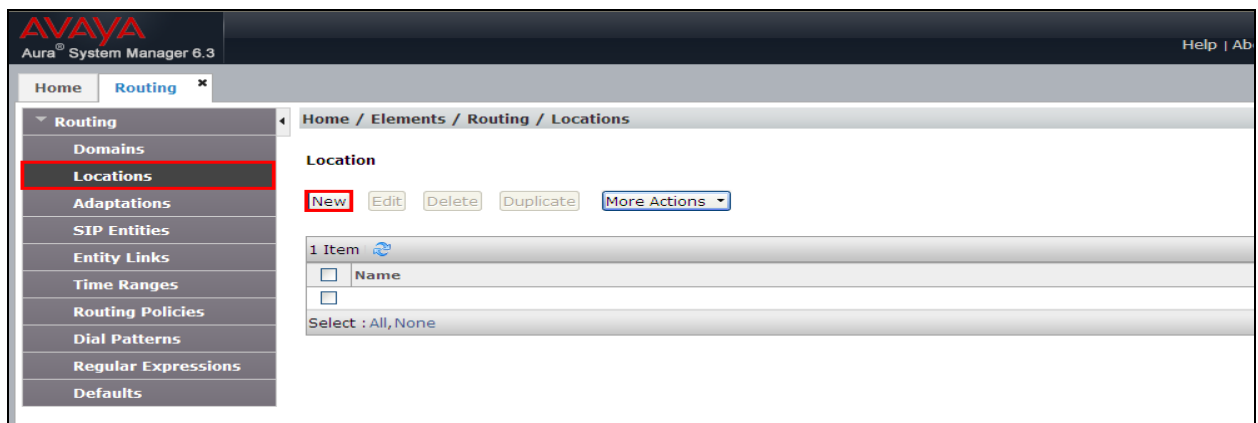


Enter the name of the domain note this was referenced in **Section 6**. The **Type** should be **sip**. Click on **Commit** once done.



## 6.3. Configure Location

Select **Locations** from the left window and select **New** from the main window.



Enter a suitable name for the location and scroll down to the bottom of the page and enter the IP addresses associated with the location in the case there are two ranges **10.10.40.x** and **192.168.50.x** and click on **Add**. Once completed, click on **Commit** to continue.

**AVAYA**  
Aura® System Manager 6.3

Home / Elements / Routing / Locations

**Location Details** Commit Cancel

**General**

\* Name: DevConnectPG63

Notes:

**Dial Plan Transparency in Survivable Mode**

Enabled: ☐

Listed Directory Number:

Associated CM SIP Entity:

\* Minimum Multimedia Bandwidth: 64 Kbit/Sec

\* Default Audio Bandwidth: 80 Kbit/sec

**Alarm Threshold**

Overall Alarm Threshold: 80 %

Multimedia Alarm Threshold: 80 %

\* Latency before Overall Alarm Trigger: 5 Minutes

\* Latency before Multimedia Alarm Trigger: 5 Minutes

**Location Pattern**

Add Remove

2 Items

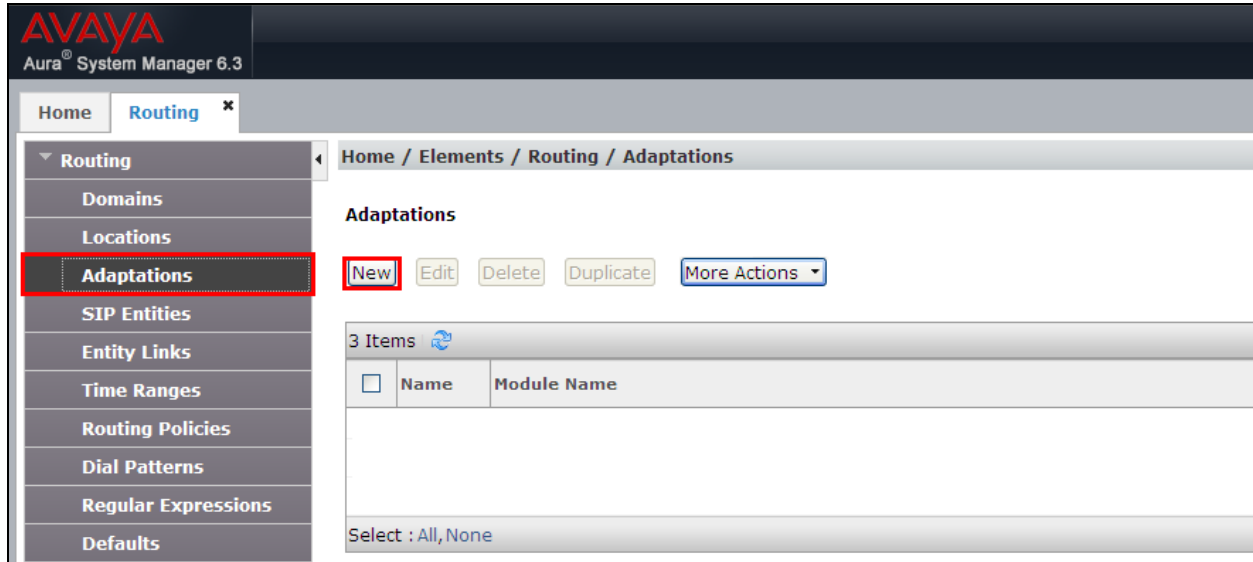
<input type="checkbox"/>	IP Address Pattern	Notes
<input checked="" type="checkbox"/>	* 10.10.40.*	
<input checked="" type="checkbox"/>	* 192.168.50.*	

Select : All, None

Commit Cancel

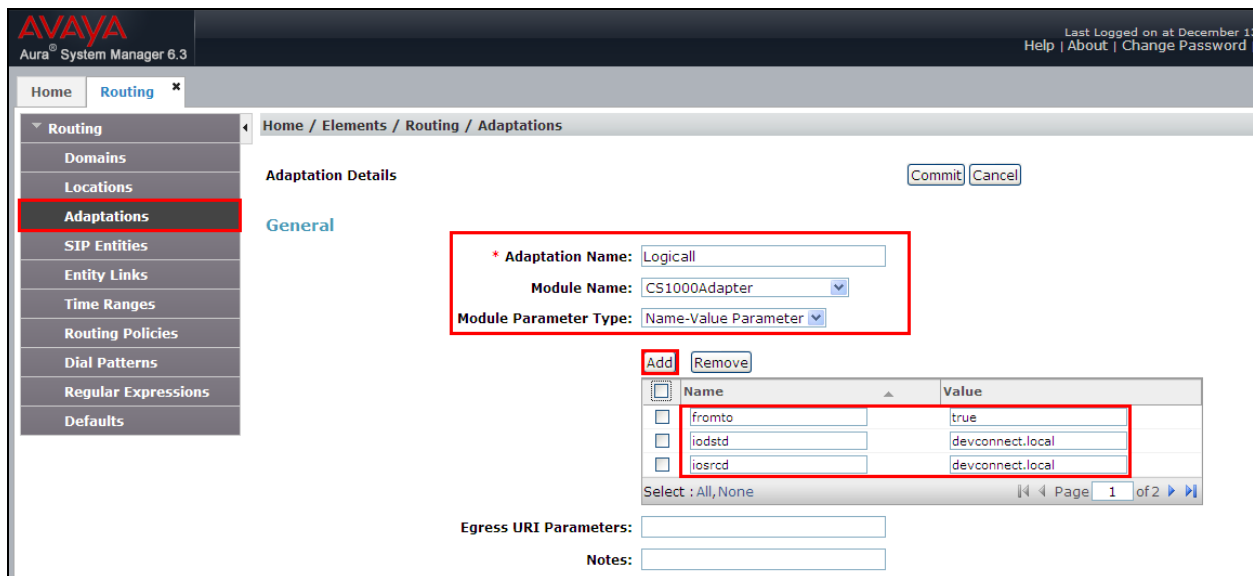
## 6.4. Configure Adaptation

To configure a new Adaptation select **Adaptations** from the left window and click on **New** from the main window.



Enter a suitable **Name** and select **DigitConversionAdapter** for the **Module Name**. Select **Name-Value Parameter** as the **Module Parameter Type**. Add the following Parameters:

- | Name   | Value    |
|--------|----------|
| fromto | true     |
| iodstd | “domain” |
| iosrcd | “domain” |



Continue to add the following Parameters:

- | Name  | Value                        |
|-------|------------------------------|
| MIME  | "No"                         |
| odstd | "Logicall IP Address"        |
| osrcd | "Session Manager IP Address" |

Click on **Commit** once completed.

AVAYA  
Aura® System Manager 6.3

Home / Elements / Routing / Adaptations

Adaptation Details **Commit** **Cancel**

General

\* Adaptation Name: Logicall

Module Name: CS1000Adapter

Module Parameter Type: Name-Value Parameter

**Add** **Remove**

Name	Value
MIME	no
odstd	10.10.40.50
osrcd	10.10.40.34

Select : All, None Page 2 of 2

Egress URI Parameters:

Notes:



## 6.5. Configure SIP Entity for Logicall IVR

Select **SIP Entities** from the left window and click on **New** in the main window.

The screenshot shows the Avaya Aura System Manager 6.3 interface. The left sidebar has a menu with 'SIP Entities' highlighted. The main window displays the 'SIP Entities' page with a 'New' button highlighted. Below the buttons is a table with 8 items. The table has columns for Name, FQDN or IP Address, Type, and Notes.

Name	FQDN or IP Address	Type	Notes
<a href="#">AAMessaging</a>	192.168.50.60	SIP Trunk	
<a href="#">ASCOMDECT1</a>	10.10.40.181	SIP Trunk	
<a href="#">CM62</a>	192.168.50.13	CM	
<a href="#">CM63VMPG</a>	10.10.40.31	CM	
<a href="#">CS1KPG1</a>	10.10.40.111	SIP Trunk	
<a href="#">CS1KPG2</a>	192.168.50.99	SIP Trunk	
<a href="#">SM63vmpg</a>	10.10.40.34	Session Manager	

Enter a suitable **Name** and ensure that the **Adaptation** that was created in **Section 7.4** is used. Enter the **Location** that was configured in **Section 7.2** and the correct **Time Zone**.

The screenshot shows the 'SIP Entity Details' form in the Avaya Aura System Manager 6.3 interface. The 'General' tab is selected. The form contains several fields, with a red box highlighting the 'Name', 'FQDN or IP Address', and 'Type' fields. Another red box highlights the 'Adaptation', 'Location', and 'Time Zone' fields. The 'Commit' button is highlighted.

**SIP Entity Details**

**General**

\* Name:

\* FQDN or IP Address:

Type:

Notes:

Adaptation:

Location:

Time Zone:

\* SIP Timer B/F (in seconds):

Credential name:

Call Detail Recording:

## 6.6. Configure Entity Link for Logical IVR

Select **Entity Link** from the left window and click on **New** in the main window.

AVAYA  
Aura System Manager 6.3

Home Routing

Home / Elements / Routing / Entity Links

Entity Links

New Edit Delete Duplicate More Actions

7 Items Filter: Enable

<input type="checkbox"/>	Name	SIP Entity 1	Protocol	Port	SIP Entity 2	DNS Override	Port	Connection Policy	Deny New Service	Notes
<input type="checkbox"/>	AAMessaging	SM63vmpg	TCP	5060	AAMessaging	<input type="checkbox"/>	5060	trusted	<input type="checkbox"/>	
<input type="checkbox"/>	ASCOMDECT1	SM63vmpg	TCP	5060	ASCOMDECT1	<input type="checkbox"/>	5060	trusted	<input type="checkbox"/>	
<input type="checkbox"/>	SM63vmpg_CM62_5061_TLS	SM63vmpg	TLS	5061	CM62	<input type="checkbox"/>	5061	trusted	<input type="checkbox"/>	
<input type="checkbox"/>	SM63vmpg_CM63VMPG_5060_TCP	SM63vmpg	TCP	5060	CM63VMPG	<input type="checkbox"/>	5060	trusted	<input type="checkbox"/>	

Select the correct **SIP Entity** that was created in **Section 7.5** and ensure that **UDP** is used as the **Protocol**. Note the **Port** is **5060**.

AVAYA  
Aura System Manager 6.3

Home Routing

Home / Elements / Routing / Entity Links

Entity Links

Commit Cancel

1 Item Filter: Enable

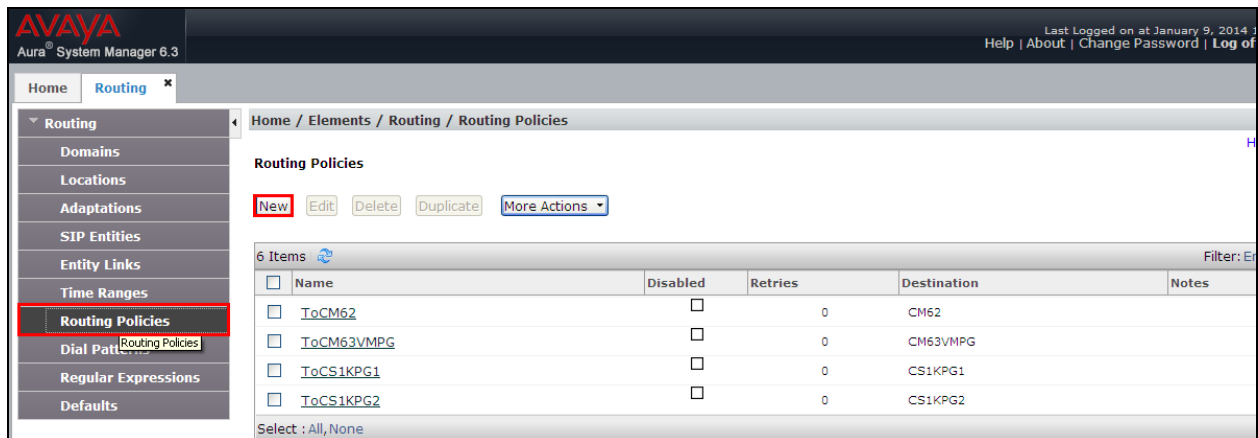
<input type="checkbox"/>	Name	SIP Entity 1	Protocol	Port	SIP Entity 2	DNS Override	Port	Connection Policy	Deny New Service	Notes
<input type="checkbox"/>	*SM63vmpg_Logical I	*SM63vmpg	UDP	*5060	*Logical IVR	<input type="checkbox"/>	*5060	trusted	<input type="checkbox"/>	

Select : All, None

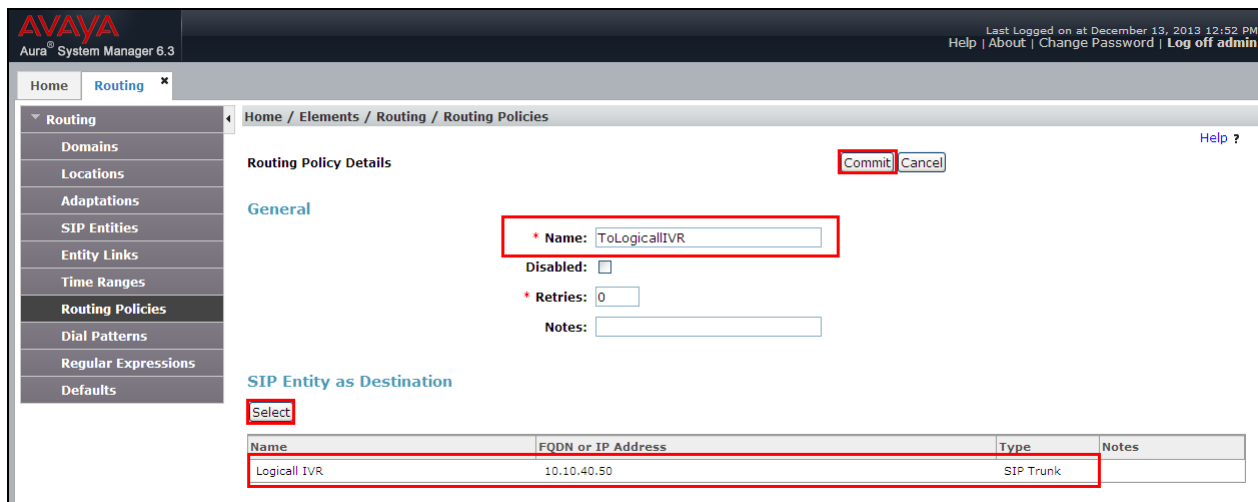
Commit Cancel

## 6.7. Configure Routing Policy for Logicall IVR

Select **Routing Policies** from the left window and click on **New** in the main window.



Enter a suitable **Name** and click on **Select** highlighted in order to associate this routing policy with a SIP Entity. Select the **Logicall** SIP Entity created in **Section 7.5** (not shown) and click on **Commit** when done.



## 6.8. Configure Dial Pattern for Logically IVR

In order to route calls to the InnLine IP, a dial pattern is created pointing to the SIP Entity. Select **Dial Patterns** from the left window and click on **New** in the main window.

The screenshot shows the Avaya Aura System Manager 6.3 interface. The left sidebar contains a navigation menu with options: Home, Routing, Domains, Locations, Adaptations, SIP Entities, Entity Links, Time Ranges, Routing Policies, **Dial Patterns** (highlighted with a red box), Regular Expressions, and Defaults. The main content area displays the 'Dial Patterns' list. At the top, there are buttons: 'New' (highlighted with a red box), 'Edit', 'Delete', 'Duplicate', and 'More Actions'. Below the buttons is a table with 6 items. The table columns are: Pattern, Min, Max, Emergency Call, Emergency Type, Emergency Priority, SIP Domain, and Notes. The table data is as follows:

Pattern	Min	Max	Emergency Call	Emergency Type	Emergency Priority	SIP Domain	Notes
10	4	4	<input type="checkbox"/>			devconnect.local	
2	4	4	<input type="checkbox"/>			devconnect.local	CM63
30	4	4	<input type="checkbox"/>			-ALL-	CS1KPG1
5999	4	5	<input type="checkbox"/>			-ALL-	AURA_Messaging
70	4	4	<input type="checkbox"/>			devconnect.local	CS1KPG1

At the bottom of the table, there is a 'Select' dropdown menu with options 'All, None'.

Enter the number to be routed noting that this will be the same number outlined in **Section 5.3.2**. Note the **SIP Domain** is that configured in **Section 7.2**. Click on **Add** to select the SIP Entity.

The screenshot shows the 'Dial Pattern Details' form in the Avaya Aura System Manager 6.3 interface. The left sidebar is the same as in the previous screenshot, with 'Dial Patterns' highlighted. The main content area shows the 'Dial Pattern Details' form. At the top right, there are 'Commit' (highlighted with a red box) and 'Cancel' buttons. The form is divided into two sections: 'General' and 'Originating Locations and Routing Policies'. The 'General' section contains the following fields:

- \* Pattern: 6000
- \* Min: 4
- \* Max: 4
- Emergency Call: ☐
- Emergency Priority: 1
- Emergency Type:
- SIP Domain: devconnect.local (highlighted with a red box)
- Notes: Logically

The 'Originating Locations and Routing Policies' section contains an 'Add' button (highlighted with a red box) and a 'Remove' button. Below these buttons, it says '1 Item'.

Tick on the **Originating Location** as shown below and select the **Logically** Routing Policy. Click on **Select** once complete.

Domains
Locations
Adaptations
SIP Entities
Entity Links
Time Ranges
Routing Policies
Dial Patterns
Regular Expressions
Defaults

Originating Location

Select

Cancel

Originating Location

☐ Apply The Selected Routing Policies to All Originating Locations

1 Item
Filter: Enable

<input checked="" type="checkbox"/>	Name	Notes
<input checked="" type="checkbox"/>	DevConnectPG63	

Select : All, None

Routing Policies

6 Items
Filter: Enable

<input type="checkbox"/>	Name	Disabled	Destination	Notes
<input checked="" type="checkbox"/>	Logically IVR	<input type="checkbox"/>	Logically IVR	
<input type="checkbox"/>	ToAAMessaging	<input type="checkbox"/>	AAMessaging	
<input type="checkbox"/>	ToCM62	<input type="checkbox"/>	CM62	
<input type="checkbox"/>	ToCM63VMPG	<input type="checkbox"/>	CM63VMPG	
<input type="checkbox"/>	ToCS1KPG1	<input type="checkbox"/>	CS1KPG1	

## 6.9. Configure SIP Entity Avaya Communication Server 1000E

Select **SIP Entities** from the left window and click on **New** in the main window.

The screenshot shows the Avaya Aura System Manager 6.3 interface. The left navigation pane has 'SIP Entities' highlighted. The main window displays the 'SIP Entities' list with 8 items. The 'New' button is highlighted in the top toolbar.

Name	FQDN or IP Address	Type	Notes
<a href="#">AAMessaging</a>	192.168.50.60	SIP Trunk	
<a href="#">ASCOMDECT1</a>	10.10.40.181	SIP Trunk	
<a href="#">CM62</a>	192.168.50.13	CM	
<a href="#">CM63VMPG</a>	10.10.40.31	CM	
<a href="#">CS1KPG1</a>	10.10.40.111	SIP Trunk	
<a href="#">CS1KPG2</a>	192.168.50.99	SIP Trunk	
<a href="#">SM63vmpg</a>	10.10.40.34	Session Manager	

Enter a suitable **Name** and ensure the **Location** that was configured in **Section 7.2** and the correct **Time Zone** is entered.

The screenshot shows the 'SIP Entity Details' form in the Avaya Aura System Manager 6.3 interface. The 'General' tab is selected. The 'Name' field is set to 'CS1KPG1' and the 'FQDN or IP Address' field is set to '10.10.40.111'. The 'Location' is set to 'DevConnectPG63' and the 'Time Zone' is set to 'Europe/Dublin'. The 'SIP Timer B/F (in seconds)' is set to 4. The 'Commit' button is highlighted.

**General**

\* Name: CS1KPG1

\* FQDN or IP Address: 10.10.40.111

Type: SIP Trunk

Notes:

Adaptation:

Location: DevConnectPG63

Time Zone: Europe/Dublin

\* SIP Timer B/F (in seconds): 4

Credential name:

Call Detail Recording: egress

## 6.10. Configure Entity Link Avaya Communication Server 1000E

Select **Entity Link** from the left window and click on **New** in the main window.

AVAYA  
Aura System Manager 6.3

Home Routing

Home / Elements / Routing / Entity Links

Entity Links

New Edit Delete Duplicate More Actions

7 Items Filter: Enable

<input type="checkbox"/>	Name	SIP Entity 1	Protocol	Port	SIP Entity 2	DNS Override	Port	Connection Policy	Deny New Service	Notes
<input type="checkbox"/>	AAMessaging	SM63vmpg	TCP	5060	AAMessaging	<input type="checkbox"/>	5060	trusted	<input type="checkbox"/>	
<input type="checkbox"/>	ASCOMDECT1	SM63vmpg	TCP	5060	ASCOMDECT1	<input type="checkbox"/>	5060	trusted	<input type="checkbox"/>	
<input type="checkbox"/>	SM63vmpg_CM62_5061_TLS	SM63vmpg	TLS	5061	CM62	<input type="checkbox"/>	5061	trusted	<input type="checkbox"/>	
<input type="checkbox"/>	SM63vmpg_CM63VMPG_5060_TCP	SM63vmpg	TCP	5060	CM63VMPG	<input type="checkbox"/>	5060	trusted	<input type="checkbox"/>	

Select the correct **SIP Entity** that was created in **Section 7.9** and ensure that **TCP** is used as the **Protocol**. Note the **Port** is **5060**. This was also given as the **Port** and **Protocol** in **Section 6**.

AVAYA  
Aura System Manager 6.3

Home Routing

Home / Elements / Routing / Entity Links

Entity Links

Commit Cancel

1 Item Filter: Enable

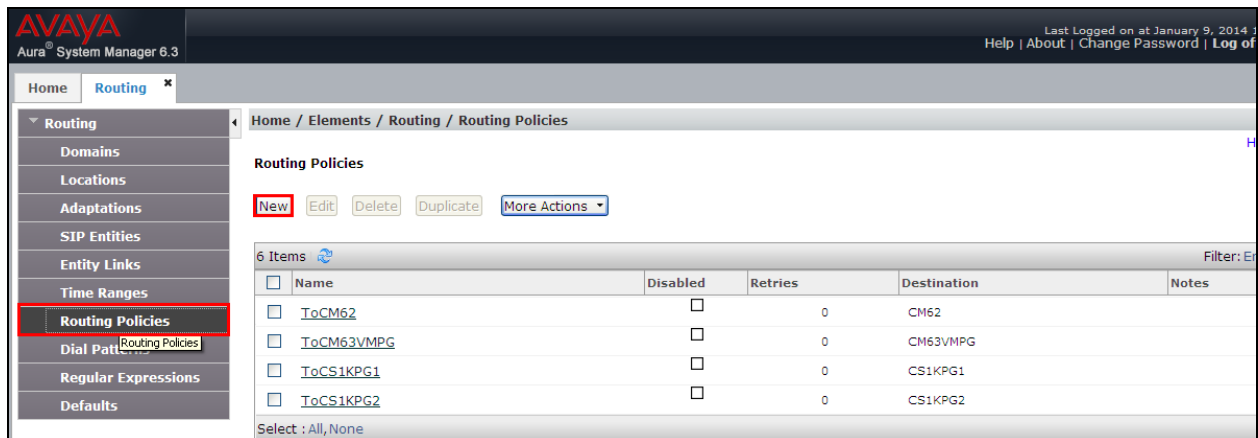
<input type="checkbox"/>	Name	SIP Entity 1	Protocol	Port	SIP Entity 2	DNS Override	Port	Connection Policy	Deny New Service	Notes
<input type="checkbox"/>	* SM63vmpg_CS1KPG	* SM63vmpg	TCP	* 5060	* CS1KPG1	<input type="checkbox"/>	* 5060	trusted	<input type="checkbox"/>	

Select : All, None

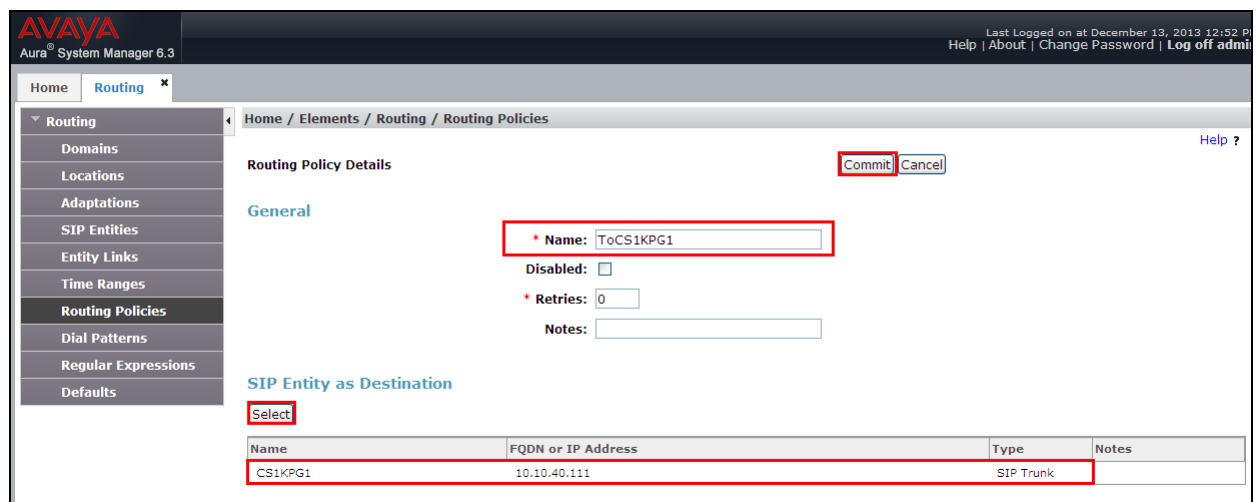
Commit Cancel

## 6.11. Configure Routing Policy Avaya Communication Server 1000E

Select **Routing Policies** from the left window and click on **New** in the main window.



Enter a suitable **Name** and click on **Select** highlighted in order to associate this routing policy with a SIP Entity. Select the **CS1000E** SIP Entity created in **Section 7.9** (not shown) and click on **Commit** when done.





## 6.12. Configure Dial Pattern for Avaya Communication Server 1000E

In order to route calls to the CS1000E a dial pattern is created pointing to the SIP Entity. Select **Dial Patterns** from the left window and click on **New** in the main window.

The screenshot shows the Avaya Aura System Manager 6.3 interface. The left sidebar contains a navigation menu with 'Dial Patterns' highlighted. The main window displays the 'Dial Patterns' list with the following data:

Pattern	Min	Max	Emergency Call	Emergency Type	Emergency Priority	SIP Domain	Notes
10	4	4	<input type="checkbox"/>			devconnect.local	
2	4	4	<input type="checkbox"/>			devconnect.local	CM63
30	4	4	<input type="checkbox"/>			-ALL-	CS1KPG1
5999	4	5	<input type="checkbox"/>			-ALL-	AURA_Messaging
70	4	4	<input type="checkbox"/>			devconnect.local	CS1KPG1

Buttons at the top include 'New', 'Edit', 'Delete', 'Duplicate', and 'More Actions'. The 'New' button is highlighted with a red box.

Enter the number to be routed noting this will be extension numbers of the CS1000E deskphones, in this case **30xx**. Note the **SIP Domain** is that configured in **Section 7.2**. Click on **Add** to select the SIP Entity.

The screenshot shows the 'Dial Pattern Details' form in the Avaya Aura System Manager 6.3 interface. The 'General' tab is active. The following fields are highlighted with red boxes:

- Pattern:** 30
- Min:** 4
- Max:** 4
- Emergency Call:** ☐
- Emergency Priority:** 1
- Emergency Type:** (empty)
- SIP Domain:** devconnect.local (dropdown menu)
- Notes:** CS1KPG1

Buttons at the top right include 'Commit' and 'Cancel'. The 'Commit' button is highlighted with a red box. Below the form, the 'Originating Locations and Routing Policies' section shows an 'Add' button highlighted with a red box.

Tick on the **Originating Location** as shown below and select the **CS1000E** Routing Policy. Click on **Select** once complete.

**Originating Location** Select Cancel

**Originating Location**

☐ Apply The Selected Routing Policies to All Originating Locations

1 Item Filter: Enable

<input checked="" type="checkbox"/>	Name	Notes
<input checked="" type="checkbox"/>	DevConnectPG63	

Select : All, None

**Routing Policies**

6 Items Filter: Enable

<input type="checkbox"/>	Name	Disabled	Destination	Notes
<input checked="" type="checkbox"/>	ToCS1KPG1	<input type="checkbox"/>	CS1KPG1	
<input type="checkbox"/>	ToAAMessaging	<input type="checkbox"/>	AAMessaging	
<input type="checkbox"/>	ToCM62	<input type="checkbox"/>	CM62	

## 7. Configuration of Maximum Network Solutions Logically IVR

The Logically IVR is preconfigured to specifications depending on the requirements of each solution. The configuration required to connect to the CS1000E is contained in the Control.ini and MYSIP.cfg files on Logically server. Once the LIVR responds to the Options messages being sent from Session Manager, the SIP trunk between the LIVR and Session Manager is completed.

All systems are configured, installed and maintained by Maximum Network Solutions.

- Ensure 'INVITE' and 'ACK' is enabled for SIP RTP re-invites.
- Any CODEC changes made to the SIP elements must also be applied to ASR server.

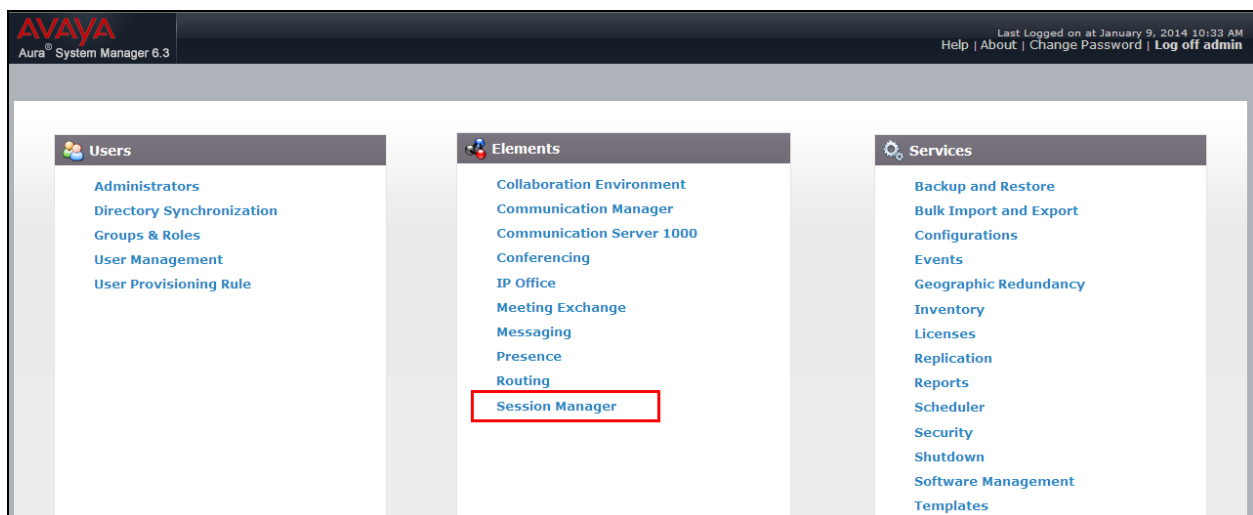
Both Control.ini and MYSIP.cfg files used in the compliance testing are included in the **Appendix B** of these Application Notes.

## 8. Verification Steps

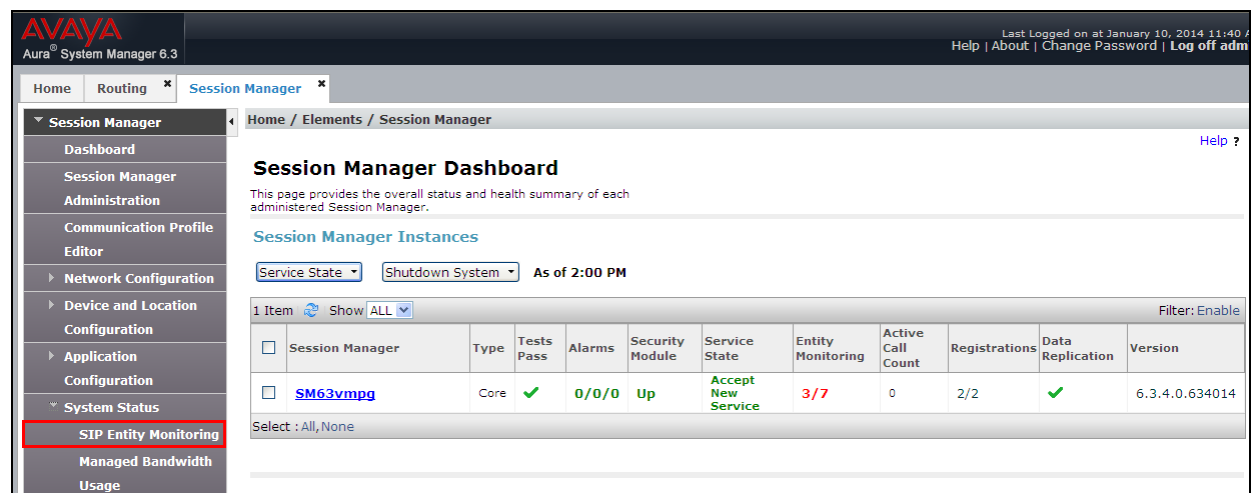
To verify a successful configuration of Logically and CS1000E a call is placed from the CS1000E to the Logically IVR application with the caller getting answered successfully, hearing clear and audible speech. A second call is made from the Logically IVR to an extension on the CS1000E by opening a demo application on Logically.

### 8.1. Verify Logically IVR SIP Entity is up

Log in to System Manager as per **Section 7.1**. From the main menu select Session Manager as shown below. Please note that on some installations Session Manager may not be setup to monitor the Logically IVR SIP Entity and in these instances the following verification test will not be required.



Navigate to **System Status** → **SIP Entity Monitoring**.



Select the **Logically IVR** SIP Entity.

The screenshot shows the 'SIP Entity Monitoring' section in the left-hand navigation menu, which is highlighted with a red box. The main content area displays a table of monitored entities. The table has columns for Session Manager, Type, and Monitored Entities (Down, Partially Up, Up, Not Monitored, Deny, Total). The first row shows 'SM63vmpg' with a 'Core' type and 11 total entities. Below the table, there is a 'Select: All, None' dropdown and a link to 'All Monitored SIP Entities'. A second table below shows 11 items, with 'Logically IVR' and 'CS1KPG2' listed. 'Logically IVR' is highlighted with a red box.

Session Manager	Type	Down	Partially Up	Up	Not Monitored	Deny	Total
<a href="#">SM63vmpg</a>	Core	6	0	5	0	0	11

SIP Entity Name
<a href="#">Logically IVR</a>
<a href="#">CS1KPG2</a>

Note that both the **Conn. Status** and **Link Status** show **UP**.

The screenshot shows the 'SIP Entity Monitoring' section in the left-hand navigation menu, which is highlighted with a red box. The main content area displays a table of monitored entities. The table has columns for Session Manager Name, SIP Entity Resolved IP, Port, Proto., Deny, Conn. Status, Reason Code, and Link Status. The first row shows 'SM63vmpg' with a '10.10.40.50' resolved IP, port 5060, UDP protocol, and 'UP' connection and link status. The 'Conn. Status' and 'Link Status' cells are highlighted with red boxes. A 'Status Details for the selected Session Manager:' box is also visible.

Session Manager Name	SIP Entity Resolved IP	Port	Proto.	Deny	Conn. Status	Reason Code	Link Status
<a href="#">SM63vmpg</a>	<a href="#">10.10.40.50</a>	5060	UDP	FALSE	UP	200 OK	UP

## 9. Conclusion

The interoperation of Logically IVR from Maximum Network Solutions with Avaya Communication Server 1000E R7.6 and Avaya Aura® Session Manager R6.3 was successful for this specific setup in order to place calls from the Logically IVR to the CS1000E and these are outlined in **Section 2.2**.

## 10. Additional References

Additional Avaya product documentation is available at <http://support.avaya.com>.

- [1] *Software Input Output Reference – Administration – Avaya Communication Server 1000, R7.6* NN43001-611
- [2] *Administering Avaya Aura® Session Manager, Release 6.3*, 03-603324

Information on Maximum Network Solutions (Maxnet) and product support visit <http://www.maxnet.co.uk>

## Appendix A

### Avaya Communication Server 1000E R7.6 - Linux Patches

Product Release: 7.65.16.00

In system patches: 0

In System service updates: 26

PATCH#	IN_SERVICE	DATE	SPECINS	REMOVABLE	NAME
2	Yes	27/08/13	NO	YES	cs1000-dmWeb-7.65.16.21-01.i386.000
3	Yes	28/08/13	NO	yes	cs1000-snmp-7.65.16.00-01.i386.000
4	Yes	28/08/13	NO	YES	cs1000-nrsm-7.65.16.00-03.i386.000
5	Yes	28/08/13	NO	YES	cs1000-oam-logging-7.65.16.01-01.i386.000
6	Yes	28/08/13	NO	yes	cs1000-cs1000WebService 6-0-7.65.16.21-00.i386.000
7	Yes	28/08/13	NO	YES	cs1000-sps-7.65.16.21-01.i386.000
8	Yes	28/08/13	NO	YES	cs1000-pd-7.65.16.21-00.i386.000
9	Yes	28/08/13	NO	YES	cs1000-shared-carrdtct-7.65.16.21-01.i386.000
10	Yes	28/08/13	NO	YES	cs1000-shared-tpselect-7.65.16.21-01.i386.000
11	Yes	28/08/13	NO	YES	cs1000-emWebLocal 6-0-7.65.16.21-01.i386.000
12	Yes	28/08/13	NO	yes	cs1000-dbcom-7.65.16.21-00.i386.000
13	Yes	28/08/13	NO	YES	cs1000-csmWeb-7.65.16.21-05.i386.000
14	Yes	28/08/13	NO	YES	cs1000-shared-xmsg-7.65.16.21-00.i386.000
15	Yes	28/08/13	NO	YES	cs1000-vtrk-7.65.16.21-29.i386.000
16	Yes	28/08/13	NO	YES	cs1000-tps-7.65.16.21-05.i386.000
17	Yes	28/08/13	NO	YES	cs1000-mscAnnc-7.65.16.21-02.i386.001
18	Yes	28/08/13	NO	YES	cs1000-mscAttn-7.65.16.21-04.i386.001
19	Yes	28/08/13	NO	YES	cs1000-mscConf-7.65.16.21-02.i386.001
20	Yes	28/08/13	NO	YES	cs1000-mscMusc-7.65.16.21-02.i386.001
21	Yes	28/08/13	NO	YES	cs1000-mscTone-7.65.16.21-03.i386.001
22	Yes	28/08/13	NO	YES	cs1000-bcc-7.65.16.21-21.i386.000
23	Yes	28/08/13	NO	YES	cs1000-Jboss-Quantum-7.65.16.21-3.i386.000
24	Yes	28/08/13	NO	YES	cs1000-emWeb 6-0-7.65.16.21-06.i386.000
25	Yes	10/12/13	NO	yes	cs1000-cs-7.65.P.100-01.i386.001
26	Yes	10/12/13	YES	yes	cs1000-linuxbase-7.65.16.21-08.i386.000
27	Yes	10/12/13	NO	YES	cs1000-patchWeb-7.65.16.21-06.i386.0

## Avaya Communication Server 1000E R7.6 - Call Server Patches

VERSION 4121  
RELEASE 7  
ISSUE 65 P +  
DepList 1: core Issue: 01 (created: 2013-06-14 03:54:33 (est))

### IN-SERVICE PEPS

PAT#	CR #	PATCH REF #	NAME	DATE	FILENAME	SPECINS
000	wi01052968	ISS1:1OF1	p32540_1	11/12/2013	p32540_1.cpl	NO
001	wi01045058	ISS1:1OF1	p32214_1	11/12/2013	p32214_1.cpl	NO
002	wi01085855	ISS1:1OF1	p32658_1	11/12/2013	p32658_1.cpl	NO
003	wi01053314	ISS1:1OF1	p32555_1	11/12/2013	p32555_1.cpl	NO
004	wi01060382	iss1:1of1	p32623_1	11/12/2013	p32623_1.cpl	YES
005	wi01070580	ISS1:1OF1	p32380_1	11/12/2013	p32380_1.cpl	NO
006	wi01067822	ISS1:1OF1	p32466_1	11/12/2013	p32466_1.cpl	YES
007	wi01061481	ISS1:1OF1	p32382_1	11/12/2013	p32382_1.cpl	NO
008	wi01072032	ISS1:1OF1	p32448_1	11/12/2013	p32448_1.cpl	NO
009	wi01022599	ISS1:1OF1	p32080_1	11/12/2013	p32080_1.cpl	NO
010	wi01035976	ISS1:1OF1	p32173_1	11/12/2013	p32173_1.cpl	NO
011	wi01065922	ISS1:1OF1	p32516_1	11/12/2013	p32516_1.cpl	NO
012	wi01055480	ISS1:1OF1	p32712_1	11/12/2013	p32712_1.cpl	NO
013	wi01041453	ISS1:1OF1	p32587_1	11/12/2013	p32587_1.cpl	NO
014	wi01078723	ISS1:1OF1	p32532_1	11/12/2013	p32532_1.cpl	NO
015	WI0110261	ISS1:1OF1	p32758_1	11/12/2013	p32758_1.cpl	NO
016	wi01064599	iss1:1of1	p32580_1	11/12/2013	p32580_1.cpl	NO
017	wi01048457	ISS1:1OF1	p32581_1	11/12/2013	p32581_1.cpl	NO
018	wi01072027	ISS1:1OF1	p32689_1	11/12/2013	p32689_1.cpl	NO
019	wi01059388	iss1:1of1	p32628_1	11/12/2013	p32628_1.cpl	NO
020	wi01074003	ISS1:1OF1	p32421_1	11/12/2013	p32421_1.cpl	NO
021	wi00933195	ISS1:1OF1	p32491_1	11/12/2013	p32491_1.cpl	NO
022	wi00996734	ISS1:1OF1	p32550_1	11/12/2013	p32550_1.cpl	NO
023	wi01065118	ISS1:1OF1	p32397_1	11/12/2013	p32397_1.cpl	NO
024	wi01063864	ISS1:1OF1	p32410_1	11/12/2013	p32410_1.cpl	YES
025	wi01072023	ISS1:1OF1	p32130_1	11/12/2013	p32130_1.cpl	YES
026	wi01075359	ISS1:1OF1	p32671_1	11/12/2013	p32671_1.cpl	NO
027	wi01080753	ISS1:1OF1	p32518_1	11/12/2013	p32518_1.cpl	NO
028	wi01070473	ISS1:1OF1	p32413_1	11/12/2013	p32413_1.cpl	NO
029	wi01075355	ISS1:1OF1	p32594_1	11/12/2013	p32594_1.cpl	NO
030	wi01071379	ISS1:1OF1	p32522_1	11/12/2013	p32522_1.cpl	NO
031	wi01070756	ISS1:1OF1	p32444_1	11/12/2013	p32444_1.cpl	NO
032	wi01075353	ISS1:1OF1	p32613_1	11/12/2013	p32613_1.cpl	NO
033	wi01062607	ISS1:1OF1	p32503_1	11/12/2013	p32503_1.cpl	NO
034	wi01068851	ISS1:1OF1	p32439_1	11/12/2013	p32439_1.cpl	NO
035	wi01075352	ISS1:1OF1	p32603_1	11/12/2013	p32603_1.cpl	NO
036	wi01092300	ISS1:1OF1	p32692_1	11/12/2013	p32692_1.cpl	NO
037	wi01063263	ISS1:1OF1	p32573_1	11/12/2013	p32573_1.cpl	NO
038	wi01087528	ISS1:1OF1	p32700_1	11/12/2013	p32700_1.cpl	NO
039	wi01055300	ISS1:1OF1	p32543_1	11/12/2013	p32543_1.cpl	NO
040	wi01039280	ISS1:1OF1	p32423_1	11/12/2013	p32423_1.cpl	NO
041	wi01068669	ISS1:1OF1	p32333_1	11/12/2013	p32333_1.cpl	NO
042	wi01069441	ISS1:1OF1	p32097_1	11/12/2013	p32097_1.cpl	NO
043	wi01058621	ISS1:1OF1	p32339_1	11/12/2013	p32339_1.cpl	NO
044	wi01032756	ISS1:1OF1	p32673_1	11/12/2013	p32673_1.cpl	NO
045	wi01070465	iss1:1of1	p32562_1	11/12/2013	p32562_1.cpl	NO
046	wi01053920	ISS1:1OF1	p32303_1	11/12/2013	p32303_1.cpl	NO
047	wi00897254	ISS1:1OF1	p31127_1	11/12/2013	p31127_1.cpl	NO
048	wi01057403	ISS1:1OF1	p32591_1	11/12/2013	p32591_1.cpl	NO
049	wi01066991	ISS1:1OF1	p32449_1	11/12/2013	p32449_1.cpl	NO
050	wi01094305	ISS1:1OF1	p32640_1	11/12/2013	p32640_1.cpl	NO



051	wi01058359	ISS1:1OF1	p32331_1	11/12/2013	p32331_1.cpl	NO
052	wi01047890	ISS1:1OF1	p32697_1	11/12/2013	p32697_1.cpl	NO
053	wi01060241	ISS1:1OF1	p32381_1	11/12/2013	p32381_1.cpl	NO
054	wi01034307	ISS1:1OF1	p32615_1	11/12/2013	p32615_1.cpl	NO
055	wi01052428	ISS1:1OF1	p32606_1	11/12/2013	p32606_1.cpl	NO
056	wi00884716	ISS1:1OF1	p32517_1	11/12/2013	p32517_1.cpl	NO
057	wi01070468	iss1:1of1	p32418_1	11/12/2013	p32418_1.cpl	NO
058	wi01091447	ISS1:1OF1	p32675_1	11/12/2013	p32675_1.cpl	NO
059	wi01068042	ISS1:1OF1	p32669_1	11/12/2013	p32669_1.cpl	NO
060	wi01061483	ISS1:1OF1	p32359_1	11/12/2013	p32359_1.cpl	NO
061	wi01065125	ISS1:1OF1	p32416_1	11/12/2013	p32416_1.cpl	NO
062	wi01056633	ISS1:1OF1	p32322_1	11/12/2013	p32322_1.cpl	NO
063	wi01070474	iss1:1of1	p32407_1	11/12/2013	p32407_1.cpl	NO
064	wi01053597	ISS1:1OF1	p32304_1	11/12/2013	p32304_1.cpl	NO
065	wi01070471	ISS1:1OF1	p32415_1	11/12/2013	p32415_1.cpl	NO
066	wi01025156	ISS1:1OF1	p32136_1	11/12/2013	p32136_1.cpl	NO
067	wi01088775	ISS1:1OF1	p32659_1	11/12/2013	p32659_1.cpl	NO
068	wi01083584	ISS1:1OF1	p32619_1	11/12/2013	p32619_1.cpl	NO
069	wi01075360	iss1:1of1	p32602_1	11/12/2013	p32602_1.cpl	NO
070	wi01053195	ISS1:1OF1	p32297_1	11/12/2013	p32297_1.cpl	NO
071	wi01043367	ISS1:1OF1	p32232_1	11/12/2013	p32232_1.cpl	NO
072	wi01082456	ISS1:1OF1	p32596_1	11/12/2013	p32596_1.cpl	NO
073	wi01089519	ISS1:1OF1	p32665_1	11/12/2013	p32665_1.cpl	NO
074	wi01065842	ISS1:1OF1	p32478_1	11/12/2013	p32478_1.cpl	NO
075	wi01088585	ISS1:1OF1	p32656_1	11/12/2013	p32656_1.cpl	NO
076	wi01035980	ISS1:1OF1	p32558_1	11/12/2013	p32558_1.cpl	NO
077	wi01087543	ISS1:1OF1	p32662_1	11/12/2013	p32662_1.cpl	NO
078	wi01060826	ISS1:1OF1	p32379_1	11/12/2013	p32379_1.cpl	NO
079	wi01061484	ISS1:1OF1	p32576_1	11/12/2013	p32576_1.cpl	NO
080	wi01034961	ISS1:1OF1	p32144_1	11/12/2013	p32144_1.cpl	NO
081	wi01056067	ISS1:1OF1	p32457_1	11/12/2013	p32457_1.cpl	NO
082	WI01077073	ISS1:1OF1	p32534_1	11/12/2013	p32534_1.cpl	NO
083	wi01073100	ISS1:1OF1	p32599_1	11/12/2013	p32599_1.cpl	NO
084	wi01060341	ISS1:1OF1	p32578_1	11/12/2013	p32578_1.cpl	NO
MDP>LAST SUCCESSFUL MDP REFRESH :2013-08-27 14:24:01(Local Time)						
MDP>USING DEPLIST ZIP FILE DOWNLOADED :2013-08-27 09:21:58(est)						

## D-Channel for SIP Trunks

```
>ld 22
REQ prt
TYPE adan dch 1
ADAN DCH 1
CTYP DCIP
DES SIPL
USR ISLD
ISLM 4000
SSRC 3700
OTBF 32
NASA NO
IFC SL1
CNEG 1
RLS ID 25
RCAP
MBGA NO
H323
OVLN NO
OVLS NO
```

## Route for SIP calls

```
>ld 21
REQ: prt
TYPE: rdb
CUST 0
ROUT 22
TYPE RDB
CUST 00
ROUT 22
DES SIPTRK
TKTP TIE
M911P NO
ESN NO
RPA NO
CNVT NO
SAT NO
RCLS EXT
VTRK YES
ZONE 00066
PCID SIP
CRID YES
SBWM NO
NODE 111
DTRK NO
ISDN YES
MODE ISLD
DCH 1
IFC SL1
PNI 00001
NCNA YES
NCRD YES
TRO YES
FALT NO
CTYP UKWN
INAC NO
ISAR NO
```

DAPC NO  
 MBXR NO  
 MBXOT NPA  
 MBXT 0  
 PTYP ATT  
 CNDP UKWN  
 AUTO NO  
 DNIS NO  
 DCDR NO  
 ICOG IAO  
 SRCH LIN  
 TRMB YES  
 STEP  
 ACOD 8022  
 TCPP NO  
 PII NO  
 AUXP NO  
 TARG  
 CLEN 1  
 BILN NO  
 OABS  
 INST  
 IDC NO  
 DCNO 0 \*  
 NDNO 0  
 DEXT NO  
 ANTK  
 SIGO STD  
 STYP SDAT  
 MFC NO  
 ICIS YES  
 OGIS YES  
 PTUT 0  
 TIMR ICF 1920  
     OGF 1920  
     EOD 13952  
     LCT 256  
     DSI 34944  
     NRD 10112  
     DDL 70  
     ODT 4096  
     RGV 640  
     GTO 896  
     GTI 896  
     SFB 3  
     PRPS 800  
     NBS 2048  
     NBL 4096  
     IENB 5  
     TFD 0  
     RTD 12  
     VSS 0  
     VGD 6  
     EESD 1024  
 SST 5 0  
 DTD NO  
 SCDT NO  
 2 DT NO  
 NEDC ORG  
 FEDC ORG  
 CPDC NO

DLTN NO  
HOLD 02 02 40  
SEIZ 02 02  
SVFL 02 02  
DRNG NO  
CDR NO  
NATL YES  
SSL  
CFWR NO  
IDOP NO  
VRAT NO  
MUS NO  
PANS YES  
MANO NO  
FRL 0 0  
FRL 1 0  
FRL 2 0  
FRL 3 0  
FRL 4 0  
FRL 5 0  
FRL 6 0  
FRL 7 0  
OHQ NO  
OHQT 00  
CBQ NO  
AUTH NO  
TDET NO  
TTBL 0  
ATAN NO  
OHTD NO  
PLEV 2  
OPR NO  
ALRM NO  
ART 0  
PECL NO  
DCTI 0  
TIDY 8022 22  
ATTR NO  
TRRL NO  
SGRP 0  
CCBA NO  
ARDN NO  
CTBL 0  
ANIE 0  
CAC\_CIS 3  
AACR NO

## Trunk channel

```
>ld 20
REQ: prt
TYPE: tn
TYPE TNB
TN 100 0 3 0
DES SIPTRK
TN 100 0 03 00 VIRTUAL
TYPE IPTI
CDEN 8D
CUST 0
XTRK VTRK
ZONE 00066
TIMP 600
BIMP 600
AUTO_BIMP NO
NMUS NO
TRK ANLG
NCOS 0
RTMB 22 1
CHID 11
TGAR 0
STRI/STRO IMM IMM
SUPN YES
AST NO
IAPG 0
CLS UNR DIP CND ECD WTA LPR APN THFD XREP SPCD MSBT
P10 NTC MID
TKID
AACR NO
DATE 27 AUG 2013
```

## Appendix B

### Maxnet Logically Control.ini

```
[General]
LogFile=/export/logically/avaya_test_com/AvayaComTest.log
Speech_Main_Path=/export/logically/avaya_test_com/wav/main/

[BlindTransfer]
BlindTransferDefault=sip:2600@47.166.92.207:5060

[SupervisedTransfer]
SupervisedTransferOriginator1=5060
SupervisedTransfer1=sip:2600@47.166.92.207:5060

[Testing]
TestNos_Enabled=no
TestCLI=0297949600
TestDNIS=2907

[SpeechResources]
RTPDestIpAddress=47.166.92.15
RTPAltDestIpAddress=N/A
RTPSrcIpAddress=47.166.92.16
RTPSrcPort_ASR=2002
RTPSrcPort_TTS=2100

[CmdRequestListener]
CmdReqListenerPort=30007

* Trace to File - Set to FALSE - Print Screen.
* Debug Notes - Trace_Status - Should be FALSE or TRUE *
* Debug Notes - Trace Level - DEBUG, ERROR, EXTENDED *
[Debug]
TraceToFile=/export/logically/avaya_test_com/AvayaComTest.log
TraceLevel=EXTENDED
```

## Maxnet Logicall MYSIP.cfg

```
#-----
#
#                               General Demo Parameters
#-----
general.board          = 0  # MNS board number to use with ADI/VCE
general.slot           = 1  # MNS timeslot to use with ADI/VCE and SIP-NCC
general.stream         = 0
general.protocol       = sip0
general.autoStart      = 0
general.autoRelease    = 0
general.autoSDP        = 1

#-----
#
#                               SIP Parameters
#-----
sip.from               = sip:6000@47.166.92.15:5060
sip.registrar          =
sip.contact             =

#-----
#
#                               SIP-SDP Parameters
#-----

sip.sdp.connection.networkType = IN
sip.sdp.connection.addressType = IP4
sip.sdp.connection.address     = 47.166.92.16  # IP address of CG board as
                                                # configured in CG cfg file
sip.sdp.connection.port       = 8004          # UDP port on CG to use
sip.sdp.origin.userName      = nmsSip
sip.sdp.origin.sessionId     = 01234567890
sip.sdp.origin.version       = 0987654321
sip.sdp.origin.networkType   = IN
sip.sdp.origin.addressType    = IP4
sip.sdp.origin.address       = 47.166.92.16   # IP address of CG board
sip.auth.user                =
sip.auth.password            =

#-----
#
#                               MSPP Parameters -- Use Fusion, not HMP
#-----
mspp.hmp                = 0

mspp.nomedia             = 0
mspp.slot                = 30

#-----
#
#                               Voice Play Parameters
#-----
voice.play.file          = play.vox
voice.play.type          = 2
voice.play.encoding      = 10

#-----
#
#                               Voice Record Parameters
#-----
voice.record.file        = record.vox
voice.record.type        = 0
voice.record.encoding    = 2
```

---

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