

## Avaya Solution & Interoperability Test Lab

# Application Notes for configuring Avaya Aura® Communication Manager R6.2 and Avaya Aura® Application Enablement Services R6.2 with VoxSpectrum DGVox v8.1 using Service Observe – Issue 1.1

#### **Abstract**

These Application Notes describe the configuration steps required for VoxSpectrum DGVox to interoperate with Avaya Aura® Communication Manager using Avaya Aura® Application Enablement Services. VoxSpectrum DGVox is a call recording solution. In the compliance testing, VoxSpectrum DGVox used the Telephony Services Application Programming Interface from Avaya Aura® Application Enablement Services to monitor stations on Avaya Aura® Communication Manager, and used the Service Observe feature via the Avaya Aura® Application Enablement Services Device, Media, and Call Control interface to capture the media associated with the monitored stations for call recording.

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.

### 1. Introduction

These Application Notes describe the configuration used to enable VoxSpectrum DGVox to interoperate with Avaya Aura® Communication Manager, Avaya Aura® Application Enablement Services, and Avaya Aura® Session Manager. VoxSpectrum DGVox offers various methods of voice recording. For the purpose of the tests described by these Application Notes, the Service Observe feature was used.

# 2. General Test Approach and Test Results

The general test approach was to validate correct recording of calls in a variety of call handling scenarios and recovery from network interruption. Parties involved in calls, clarity of recording and accurate call times and durations were verified. The resumption of call recording following outages of various components of the solution was also checked.

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.

# 2.1. Interoperability Compliance Testing

Interoperability compliance testing consisted of the successful, clear and accurate recording of both monitored and unmonitored extensions, as well as recovery from failure in the following scenarios:

- Internal calls called/calling party ends call
- Calls between networked PBX's inbound/outbound called party/calling party ends call
- PSTN Calls inbound/outbound called party/calling party ends call
- Hold/Retrieve
- Supervised/Unsupervised Transfer
- Conference
- Call Forwarding
- Hunt Group Calls
- Bridged Appearance answered/placed by bridged appearance
- Calls gone to cover
- PBX restart recovery
- DGVox network recovery
- DGVox power outage recovery

#### 2.2. Test Results

All test cases passed successfully with the following observations:

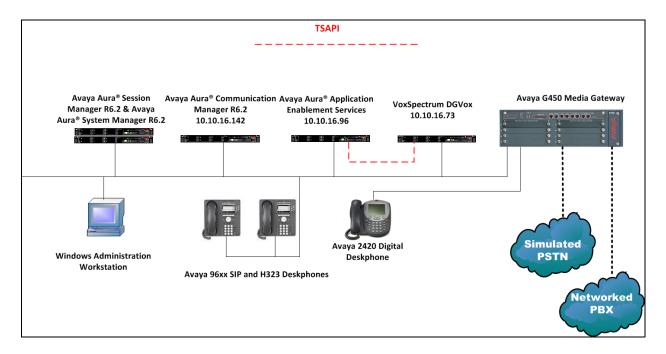
- Where a call is placed from recorded H.323 station A over an H.323 trunk to a networked PBX, the recording fails. This can be rectified by disabling shuffling on the H.323 trunk.
- Where a call is placed from recorded station A over an H.323 trunk to a networked PBX, and A puts the call on hold, only the part of the call after the call is retrieved is recorded. This can be rectified by disabling shuffling on the H.323 trunk.
- Where a call is placed to or from a recorded SIP station C and C performs hold and retrieve activities, the second recorded leg of the call shows outgoing with no dialed digits.
- Where a call is placed from station A to station C and C conferences SIP station D the channel status does not show an active call for C.
- Where A is forwarded to unrecorded station E and C calls A, station E does not appear on the recording data.
- Where a bridged appearance is configured on E for A or C and a call is placed to A or C and answered using the bridged appearance of A or C on E, the recording data does not contain any dialed digits or caller ID.
- Where a bridged appearance for recorded SIP station D is configured on E and a call is placed from the bridged appearance for D on E the call recording data shows as incoming with no dialed digits or caller ID.
- Where a call is placed between A and C and the LAN cable is disconnected and after 5 minutes reconnected to the DGVox Server, the legs of the 2 call prior to disconnection appear as outgoing in the recording data. After reconnection the recording is resumed correctly and when the call ends the recording data is accurate.

# 2.3. Support

Support for DGVox is available at: support@voxspectrum.com

# 3. Reference Configuration

An Avaya S8800 Server running Avaya Aura® Communication Manager R6.2 serving H.323 endpoints with an Avaya G450 Media Gateway was configured along with Avaya Aura® Session Manager R6.2 hosted on an Avaya S8800 Server providing SIP endpoints. VoxSpectrum DGVox was configured on the same IP network for connection to Avaya Aura® Application Enablement Services over TSAPI.



Avaya Aura® Communication Manager, Avaya Aura® Session Manager and Avaya Aura® Application Enablement Services with VoxSpectrum DGVox Solution

# 4. Equipment and Software Validated

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

<b>Equipment/Software</b>	Release/Version
Avaya Aura® Communication Manager	R6.2 SP4 build R016x.02.0.823.0-20199
running on Avaya S8800 Server	
Avaya Aura® Session Manager running on	R6.2 SP3
Avaya S8800 Server	
Avaya Aura® Application Enablement	R6.2
Services	
Avaya G450 Media Gateway	31.22.0
• MM710	• HW5 FW22
• MM712	• HW7 FW14
Avaya 9630 IP Deskphone	• H.323 S3.1 SP5
	• SIP 2.6 SP9
Avaya 2420 Digital Deskphone	2420 Rel 6.00 HWT=51H HWV=1
	FWV=6
VoxSpectrum DGVox	• v8.1
	<ul> <li>Avaya Application Enablement</li> </ul>
	Services TSAPI Client 4.2.474

# 5. Configure Avaya Aura® Communication Manager

The configuration and verification operations illustrated in this section were all performed using the Communication Manager System Administration Terminal (SAT). It is assumed that the relevant dialplan, hunt groups, stations, trunks and call routing have been configured. The connection from Communication Manager to Session Manager is not specific to the test environment and is therefore not detailed below.

The information provided in this section describes the configuration of Communication Manager for this solution. For all other provisioning information such as installation and configuration, please refer to the product documentation in **Section 10**.

# **5.1. Configure AEServices**

An AE Services link must be established between Communication Manager and Application Enablement Services. Enter the command **change node-names ip** and enter the node **Name** and **IP Address** for Application Enablement Services in this case **10.10.16.96**. Take a note of the **procr** node **Name** and **IP Address**, in this case **10.10.16.142**.

change node-names	ip			Page	1 of	2
		IP NODE	NAMES			
Name	IP Address					
procr	10.10.16.142					
CM521	10.10.16.23					
Gateway	10.10.16.1					
IPbuffer	10.10.16.184					
Intuition	10.10.16.51					
MedPro	10.10.16.32					
Presence	10.10.16.83					
RDTT	10.10.16.185					
SESMNGR	10.10.16.44					
SM1	10.10.16.43					
SM61	10.10.16.201					
default	0.0.0.0					
aesserver62	10.10.16.96					

In order for Communication Manager to establish a connection to Application Enablement Services, administer the CTI Link as shown below. Using the **add cti-link next** command specify an available **Extension** number, set the **Type** as **ADJ-IP**, which denotes that this is a link to an IP connected adjunct, and name the link for easy identification, in this instance, the nodename is used.

add cti-li	nk next			Page	1	of	3
		CTI	LINK				
CTI Link:	1						
Extension:	5899						
Type:	ADJ-IP						
						COR:	1
Name:	aesserver62						

Using the command **change ip-services**, configure IP-Services using **AESVCS** as the **Service Type** enter the **procr** node name as noted above as the **Local Node** 

change ip-	services				Page 1 of	4
Service Type	Enabled	Local Node	IP SERVICES Local Port	Remote Node	Remote Port	
AESVCS	У	procr	8765			

On **Page 4**, set the **AE Services Server** node-name and the **Password** that Application Enablement Services will use to authenticate with Communication Manager.

change ip-se	rvices			Page	<b>4</b> of	4
AE Services Administration						
Server ID		Password	Enabled	Status		
1:	Server aesserver62	Avayapassword1	У	in use		

## 5.2. Configure Class of Restriction

A class of restriction must be configured in order that stations can be service observed and/or be service observers. For the purpose of the compliance test both the service observe stations and the recorded stations used COR 1. Enter the command **change cor 1** and configure **Can Be Service Observed** and **Can Be A Service Observer** to y.

```
change cor 1
                                                                                                                                                                                                                             Page
                                                                                                                                                                                                                                                      1 of
                                                                                                                                                                                                                                                                           23
                                                                                                          CLASS OF RESTRICTION
                                                     COR Number: 1
                                    COR Description:
                                                                                                                                                                                                                     APLT? v
       Can Be Service Observed? y
                                                                                                                                 Calling Party Restriction: none
                               Service Observed? y

Service Observer? y

Calling Party Restriction: no Called Party Restriction: no Ca
Can Be A Service Observer? y
                                                                                                                                         Called Party Restriction: none
                            Time of Day Chart: 1
                 Restriction Override: none Facility Access Trunk Test? n
                 Restricted Call List? n
                                                                                                                                                               Can Change Coverage? n
                                                                                                                                         Fully Restricted Service? n
Access to MCT? y
Group II Category For MFC: 7
                                          Access to MCT? y
                                                                                                                                             Hear VDN of Origin Annc.? n
                                Send ANI for MFE? n
                                                                                                                                              Add/Remove Agent Skills? n
                                         MF ANI Prefix:
                                                                                                                                             Automatic Charge Display? n
Hear System Music on Hold? y PASTE (Display PBX Data on Phone)? n
                                                                                     Can Be Picked Up By Directed Call Pickup? y
                                                                                                                                Can Use Directed Call Pickup? y
                                                                                                                                Group Controlled Restriction: inactive
```

## 5.3. Configure System-Wide Service Observe Features

Enter the command change system-parameters features and on Page 11 set Service Observing: Warning Tone to n and Service Observing/SSC Allowed with Exclusion and Allow Two Observers in Same Call to y.

```
change system-parameters features
                                                              Page 11 of 19
                        FEATURE-RELATED SYSTEM PARAMETERS
CALL CENTER SYSTEM PARAMETERS
  EAS
        Expert Agent Selection (EAS) Enabled? y
       Minimum Agent-LoginID Password Length:
          Direct Agent Announcement Extension:
                                                                  Delay:
   Message Waiting Lamp Indicates Status For: station
  VECTORING
                    Converse First Data Delay: 0
                                                      Second Data Delay: 2
                                                        Pause (msec): 70
               Converse Signaling Tone (msec): 100
                     Prompting Timeout (secs): 10
                 Interflow-qpos EWT Threshold: 2
   Reverse Star/Pound Digit For Collect Step? n
         Available Agent Adjustments for BSR? n
                            BSR Tie Strategy: 1st-found
   Store VDN Name in Station's Local Call Log? n
  SERVICE OBSERVING
                                                     or Conference Tone? n
              Service Observing: Warning Tone? n
 Service Observing/SSC Allowed with Exclusion? y
            Allow Two Observers in Same Call? y
```

## 5.4. Configure Service Observe Stations

DGVox uses a pool of stations as recording extensions, these are used to service observe into stations which are configured to have their calls recorded. Enter the command **add station next** and configure a relevant **Extension**, set the **Security Code**, set the **Type** as **4624**, the **Port** as **IP** and assign an identifying **Name**. Ensure that **IP SoftPhone** is set to **y** and the **COR** is that configured in **Section 5.2**. Repeat this according to the number extensions required by DGVox. During the compliance test 4 stations were configured for this purpose, 6500 – 6503.

add station next Page **1** of STATION Extension: 6500 Lock Messages? n BCC: 0 TN: 1 Type: 4624 Security Code: 1234 Port: IP Coverage Path 1: COR: 1 Name: Recorder, 6500 Coverage Path 2: cos: 1 Hunt-to Station: STATION OPTIONS Time of Day Lock Table: Loss Group: 19 Personalized Ringing Pattern: 1 Message Lamp Ext: 6500 Speakerphone: 2-way Mute Button Enabled? y Display Language: english Survivable GK Node Name: Survivable COR: internal Media Complex Ext: Survivable Trunk Dest? y IP SoftPhone? y IP Video Softphone? n Short/Prefixed Registration Allowed: default

#### On Page 4 configure Button Assignment 8 with the service-obsrv feature.

```
change station 6500
                                                                 Page
                                                                         4 of
                                       STATION
 SITE DATA
      Room:
                                                           Headset? n
       Jack:
                                                           Speaker? n
      Cable:
                                                         Mounting: d
                                                       Cord Length: 0
      Floor:
   Building:
                                                         Set Color:
ABBREVIATED DIALING
    List1:
                                List2:
                                                            List3:
BUTTON ASSIGNMENTS
                                           7:
 1: call-appr
 2: call-appr
                                           8: serv-obsrv
                                           9:
                                          10:
 4: conf-dsp
 5:
                                          11:
 6:
                                          12:
```

# 5.5. Configure SIP Stations for CTI Control

SIP stations must be configured so they can be monitored by DGVox, enter the command **change station xxxx** where **xxxx** is a SIP extension and configure **Type of 3PCC Enabled** to **Avaya** in this instance on **Page 6**. For the purposes of the compliance test SIP stations 6002 and 6003 were configured.

```
change station 6002

STATION

SIP FEATURE OPTIONS

Type of 3PCC Enabled: Avaya

SIP Trunk: aar
```

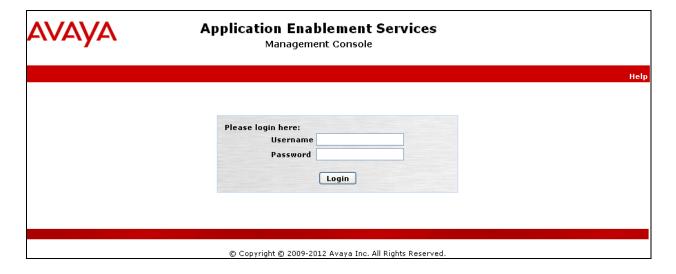
# 6. Configure Avaya Aura® Application Enablement Services Server

This section provide the procedures for configuring Application Enablement Services. The procedures include the following areas:

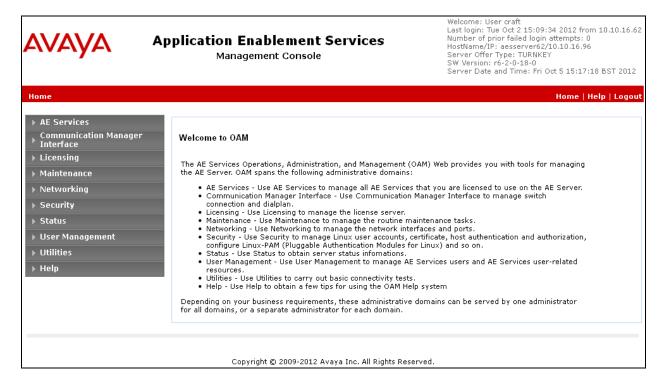
- Launch OAM interface
- Administer the Switch Connection
- Administer TSAPI Link
- Restart TSAPI Service
- Obtain Tlink name
- Administer Avaya CTI User

#### 6.1. Launch OAM Interface

Access the OAM web-based interface of AE Services, in this instance using the URL <a href="https://10.10.16.96">https://10.10.16.96</a>. The Management console is displayed. Log in using the appropriate credentials.

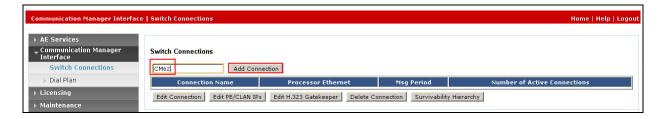


The **Welcome to OAM** screen is displayed next.

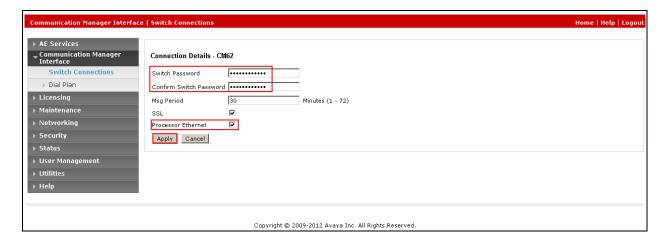


#### 6.2. Administer the Switch Connection

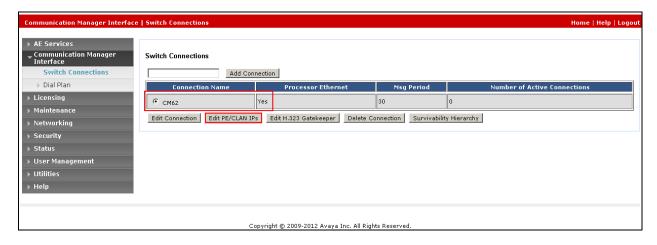
To establish the connection between Communication Manager and AE Services, click Communication Manager Interface → Switch Connections. In the field next to Add Connection enter CM62 and click on Add Connection, the following screen will be displayed.



The following screen is displayed. Complete the configuration as shown and enter the password specified in **Section 5.1** when configuring AESVCS in ip-services. Click on **Apply** when done.



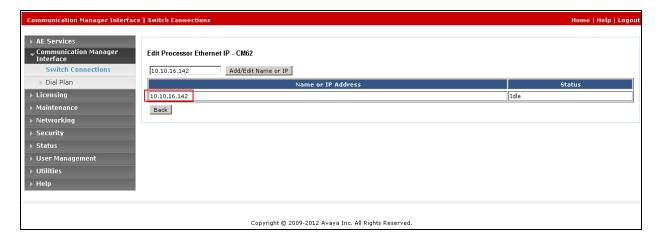
The following screen will be shown displaying the newly added switch connection, click on **Edit PE/CLAN IPs** in order to specify the IP address of the procr, as noted in **Section 5.1**.



Next to **Add name or IP**, enter the IP address of the procr as shown below.

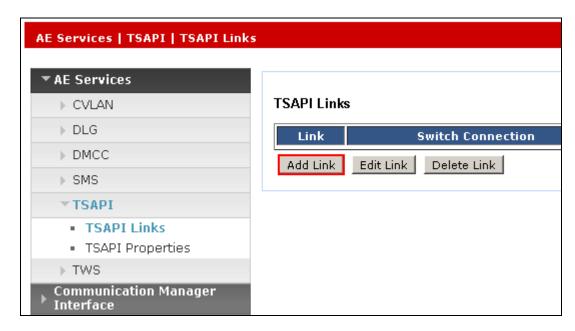


The following screen will now appear displaying the newly added IP address.

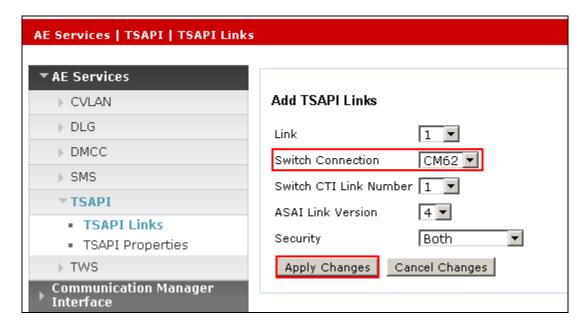


#### 6.3. Administer TSAPI Link

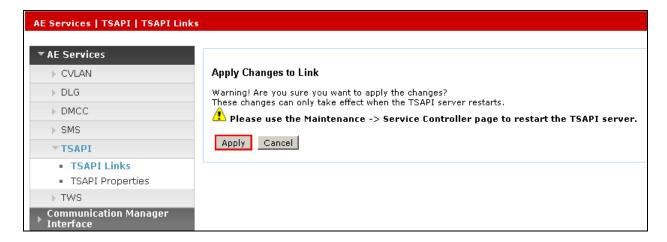
Select **AE Services** → **TSAPI → TSAPI Links** from the left pane. The **TSAPI Links** screen is displayed, click **Add Link**.



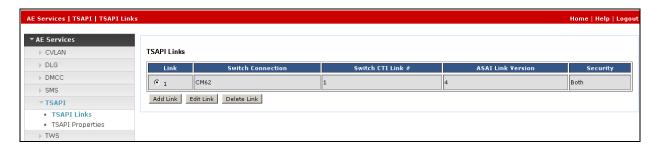
Configure the TSAPI Link using the newly configured **Switch Connection** as shown below and click **Apply Changes**.



The screen below will be displayed with instructions to restart the TSAPI Server. Click **Apply** taking note of the instructions given.

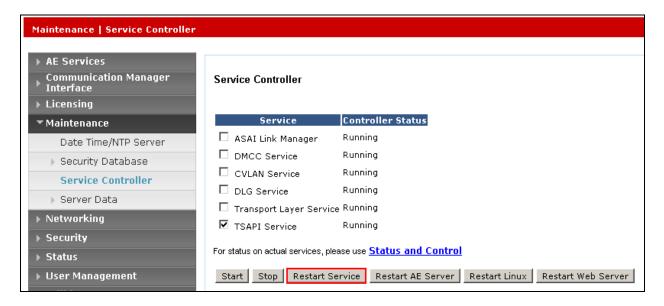


The following screen will be displayed showing the TSAPI Link.



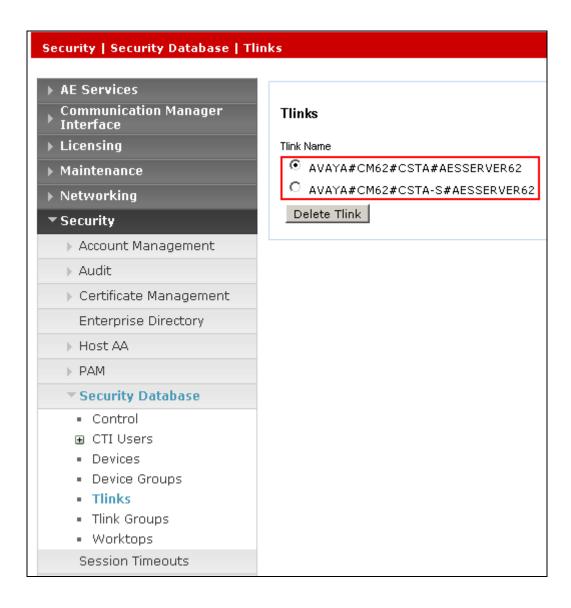
#### 6.4. Restart TSAPI Service

Select Maintenance  $\rightarrow$  Service Controller from the left pane, to display the Service Controller screen in the right pane. Check the TSAPI Service, and click Restart Service.



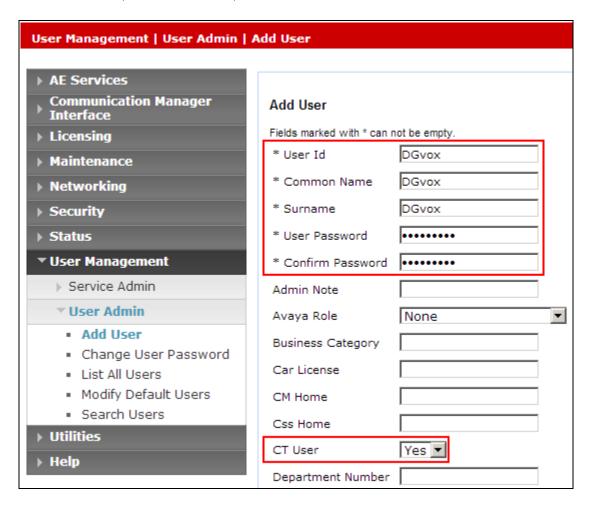
#### 6.5. Obtain Tlink Name

Select Security Security Database Tlinks from the left pane. The Tlinks screen shows a listing of the Tlink names. Locate the Tlink name associated with the relevant switch connection, which would use the name of the switch connection as part of the Tlink name.

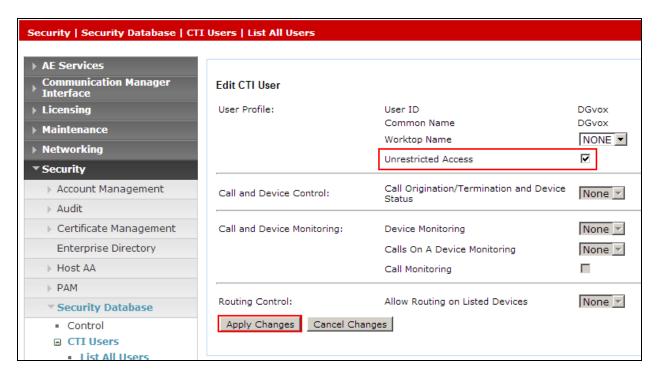


#### 6.6. Administer CTI User

In this section a CTI user is configured for DGVox to communicate with Application Enablement Services. Select **User Management** → **User Admin** → **Add User** from the left pane to display the **Add User** screen in the right pane. Enter desired values for **User Id**, **Common Name**, **Surname**, **User Password** and **Confirm Password**. For **CT User**, select **Yes** from the drop-down list. Retain the default value in the remaining fields. Click Apply at the bottom of the screen (not shown below).

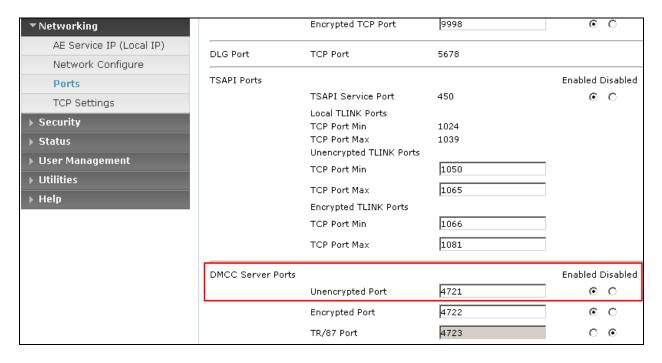


This user should be configured as an unrestricted user. Select Security  $\rightarrow$  Security Database  $\rightarrow$  CTI Users  $\rightarrow$  List All Users from the left pane, click on the radio button beside the user created above, in this case, DGvox and click Edit. Place a tick in the box next to Unrestricted Access, as shown in the image below. Click Apply Changes when done.



# 6.7. Configure Port for Unencrypted DMCC Connection

Click **Networking** → **Ports**, in the **DMCC Server Ports** section ensure that **Unencrypted Port** is **Enabled** and set to **4721**. Click Apply Changes (not shown) when done.

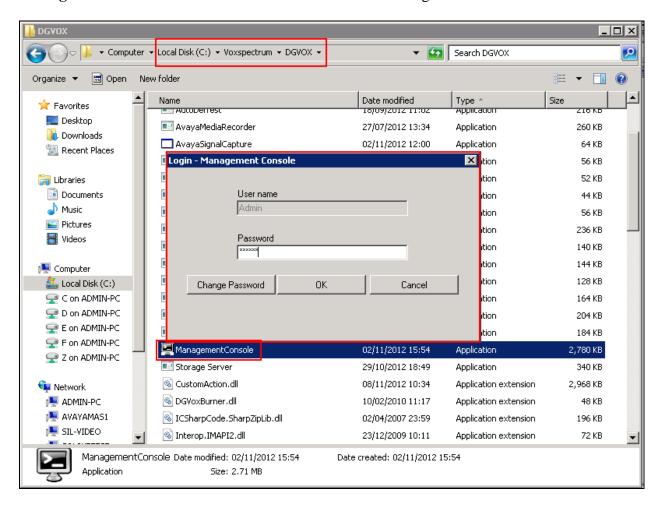


# 7. Configure VoxSpectrum DGVox

DGVox is installed and commissioned by a VoxSpectrum commissioning engineer. The following section describes the configuration necessary for interfacing with the Avaya components of the solution.

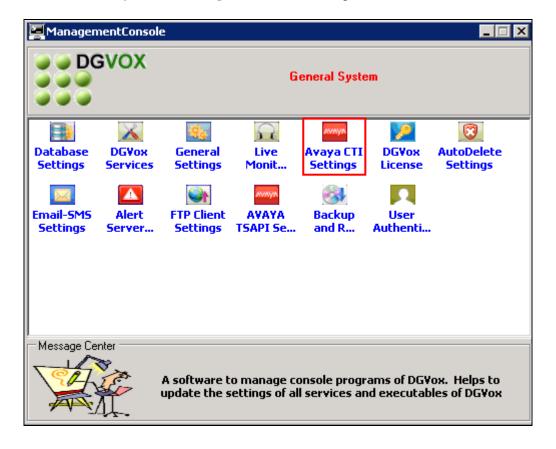
# 7.1. Login to Management Console

The initial DGVox configuration is administered using the Management Console, the default location is **c:\Voxspectrum\DGVOX\ManagementConsole.exe**. Double click on the **Management Console** icon and enter the user credentials to log in.



## 7.2. Administer CTI Connection

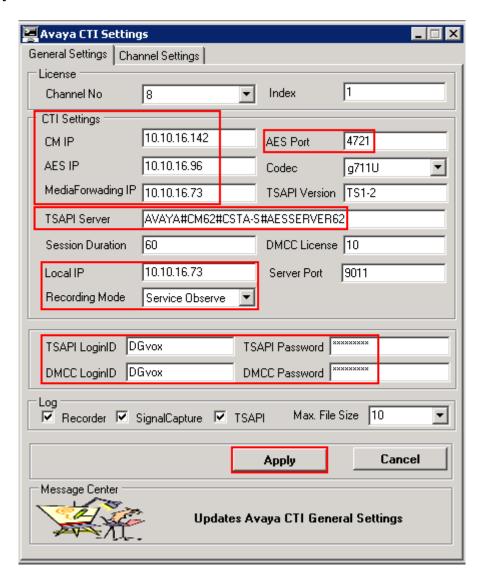
Double click on the Avaya CTI Settings icon in the Management Console



The screen below shows the configuration required to connect to DGVox to Communication Manager and Application Enablement Services and can be described as follows:

- **CM IP** enter the procr IP address
- **AES IP** enter the Application Enablement Services IP address
- MediaForwading IP enter the IP address of the DGVox server
- AES Port enter the unencrypted DMCC port number administered in Section 6.7
- TSAPI Server enter the Tlink string obtained in Section 6.5
- Local IP enter the IP address of the DGVox Server
- **Recording mode** select **Service Observe** from the drop down box
- TSAPI LoginID and DMCC LoginID enter the CTI user administered in Section 6.6
- TSAPI a Password and DMCC Password enter the CTI user password administered in Section 6.6

Click **Apply** when done.

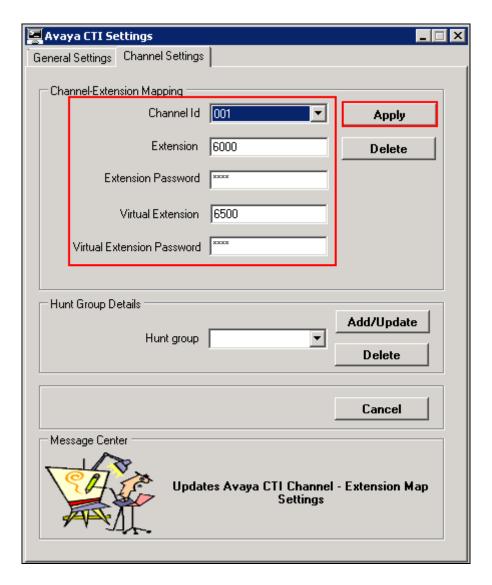


# 7.3. Configure Recording Channel Mapping

Each of the administered recording extensions are configured with a Channel ID and mapped to the station which must be recorded. Click on the **Channel Settings** tab and configure Channel Extension Mapping as follows:

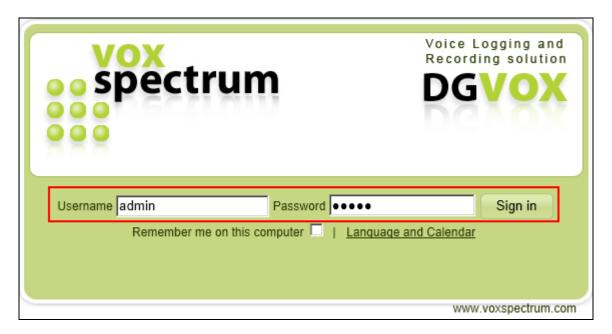
- Channel ID choose the first channel number in this case 001
- Extension enter the station number to be recorded
- Extension Password enter the Security Code configured for this station
- Virtual Extension enter the recorder station configured in **Section 5.4**
- Virtual Extension Password enter the security code for the station configured in **Section** 5.4

Click **Apply** when done and repeat the task for the remaining Channel ID and stations to be recorded.

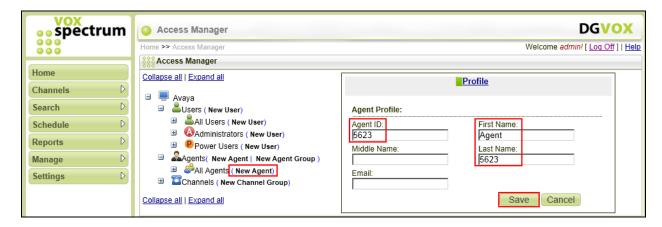


# 7.4. Add Agent Configuration

Navigate to the URL of the DGVox web interface, in this case <a href="https://10.10.16.73/DGVOX/">https://10.10.16.73/DGVOX/</a> and enter the appropriate login credentials.



Click **Manage** → **Access Manager** → **New Agent** and add the details for any Communication ACD agents to be recorded. Click **Save when done.** 

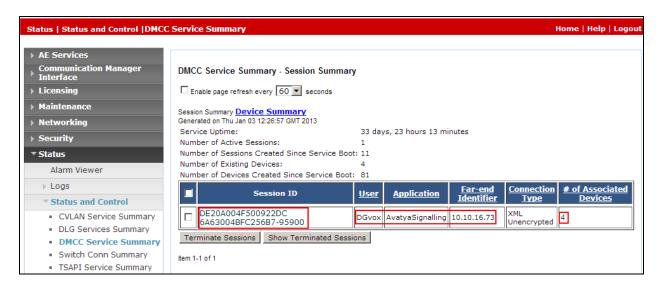


# 8. Verification Steps

The following steps can be used to verify the correct operation of the Avaya and VoxSpectrum solution.

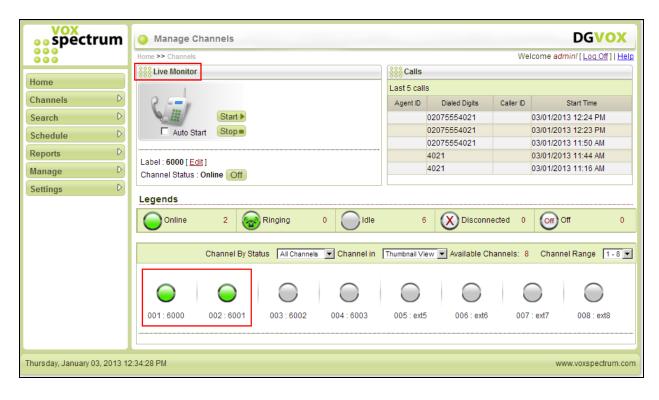
# 8.1. Verify Avaya Aura® Application Enablement Services DMCC Status

Using the Application Enablement Services web interface click Status → Status and Control → DMCC Service Summary confirm that there is an active Session ID, the User is that configured in Section 6.6, the Application is Avatya Signalling which represents the DGVox application, the Far-end Identifier is the IP address assigned to the DGVox server, and the # of Associated Devices relates to the number of service observer recorder stations configured, in this case 4.



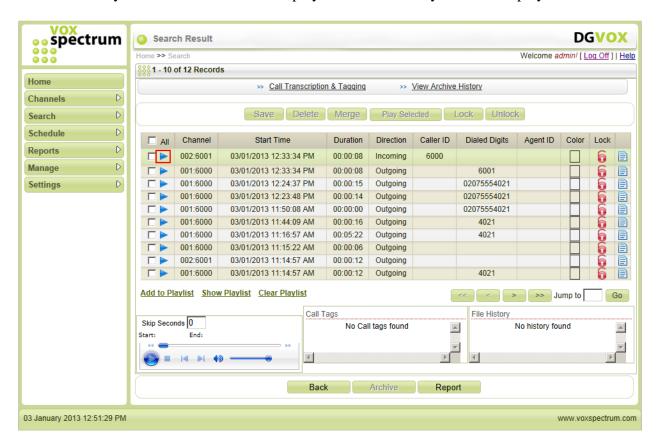
# 8.2. Verify VoxSpectrum Limited DGVox Live Monitor

From the DGVox web interface, click **Home** and confirm that the **Live Monitor** accurately displays the activity on the recorder at that time. In this case, extensions **6000** and **6001** are active and being recorded.



## 8.3. Verify VoxSpectrum Limited DGVox Search Results and Playback

From the DGVox web interface, click **Search Search** enter the search parameters required, and click **Search** (not shown). Verify that the search results accurately present the call activity. Click the symbol next to the call to be played back and verify clear audio playback is heard.



### 9. Conclusion

All test cases were executed successfully with observations in **Section 2.2**. It is not recommended to use shuffling if H.323 trunks are in use.

# 10. Additional References

This section references the product documentations that are relevant to these Application Notes.

Avaya product documentation can be found at http://support.avaya.com.

• Administering Avaya Aura® Communication Manager, Release 6.2, 03-300509, Issue 7.0 December 2012

VoxSpectrum product documentation can be obtained by using the contact details in **Section 2.3**.

• DGVox 8.1 user manual and installation manual

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