

Avaya Solution & Interoperability Test Lab

Application Notes for Mattersight Call Recording Solution with Avaya Aura® Communication Manager Using Single Step Conference with Avaya Aura® Application Enablement Services – Issue 1.0

Abstract

These Application Notes describe the configuration steps required for Mattersight Call Recording Solution to interoperate with Avaya Aura® Communication Manager using Avaya Aura® Application Enablement Services.

Mattersight Call Recording Solution is a call recording solution. In the compliance testing, Mattersight Call Recording Solution used the Device, Media, and Call Control interface from Avaya Aura® Application Enablement Services to monitor skill group and agent station extensions on Avaya Aura® Communication Manager, and to capture the media associated with the monitored agents for call recording.

Readers should pay attention to **Section 2**, in particular the scope of testing as outlined in **Section 2.1** as well as any observations noted in **Section 2.2**, to ensure that their own use cases are adequately covered by this scope and results.

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 steps required for Mattersight Call Recording Solution to interoperate with Avaya Aura® Communication Manager 7 using Avaya Aura® Application Enablement Services 7.

Mattersight Call Recording Solution is a call recording solution. In the compliance testing, Mattersight Call Recording Solution used the Device, Media, and Call Control (DMCC) interface from Avaya Aura® Application Enablement Services to monitor skill group and agent station extensions on Avaya Aura® Communication Manager, and to capture the media associated with the monitored agents for call recording.

When there is an active call on the monitored agent, Mattersight Call Recording Solution is informed of the call via event reports from the DMCC interface. Mattersight Call Recording Solution starts the call recording by using the Single Step Conference from the DMCC interface to conference a virtual IP softphone to the active call to obtain the media. The event reports are also used to determine when to stop the call recordings.

2. General Test Approach and Test Results

The feature test cases were performed both automatically and manually. Upon start of the Call Recording Solution application, the application automatically uses DMCC to register the virtual IP softphones to Communication Manager, and to request monitoring on the recording skill group and agent station extensions.

For the manual part of the testing, each call was handled manually on the agent telephone with generation of unique audio content for the recordings. Necessary user actions such as hold and reconnect were performed from the agent telephones to test the different call scenarios. The serviceability test cases were performed manually by disconnecting/reconnecting the Ethernet connection to Call Recording Solution.

The verification of tests included using the Call Recording Solution logs for proper message exchanges, and using the Retrieval application for proper logging and playback of the calls.

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

The interoperability compliance test included feature and serviceability testing.

The feature testing focused on verifying the following on Call Recording Solution:

- Use of DMCC registration services to register and un-register the virtual IP softphones.
- Use of DMCC monitoring services to monitor agent stations and virtual IP softphones.
- Use of DMCC call control service to activate Single Step Conference to conference virtual IP softphones to active call and to obtain the media for call recording.
- Proper recording, logging, and playback of calls for scenarios involving G711, G729, inbound, outbound, internal, external, ACD, non-ACD, hold, reconnect, simultaneous calls, simultaneous agents, conference, transfer and long call duration.

The serviceability testing focused on verifying the ability of the Call Recording Solution to recover from adverse conditions, such as disconnecting/reconnecting the Ethernet connection to the Call Recording Solution server.

2.2. Test Results

All test cases were executed, and the following were observations on the Call Recording Solution from the compliance testing:

- The User interface to display recordings is not part of this solution therefore the test audio files created after the call completed were played from Windows File Explorer. Call details such as DNIS, Call Type, Call Direction (meta data) was reviewed separately.
- Agents must be a member of a specific skill group in order to be recorded.
- If the agent is reskilled without using the Avaya CMS tool the agent will need to log out and log back in for the skill changes to take effect.
- During the call if a user performs actions such as pressing the Transfer button on deskphone and then pressing the Cancel or Clear button this will trigger AES to send a Call Clear event and Mattersight will stop recording the second leg of the call. The first leg of the call will continue to be recorded.
- When an agent is active on a call that is being monitored and recorded, and the agent places another call (by using transfer, conference, or hold) to a destination that is not monitored, the call to the non-monitored destination will not be recorded. Mattersight has indicated that this is working as designed.
- For a call between 2 internal agents, only1 recording will be created.
- The voice quality of a recorded G729 call is not as clear as a G.711 call.

2.3. Support

Technical support on Mattersight Call Recording Solution can be obtained through the following:

• **Phone:** 877.235.6925

• Email: support@mattersight.com

3. Reference Configuration

The Mattersight Call Recording Solution is a SaaS (Software as a service solution) offering. Mattersight will design an appropriately sized solution based on several factors to include concurrent calls, calls per second, and concurrent agents. The solution is made up of components distributed across multiple servers. The compliance test used a single server configuration shown in **Figure 1**.

There is no user interface provided for Call Recording Solution to review and playback the call recordings, to verify if Call Recording Solution properly recorded the call, manually open media file locate in folder D:\AFCRecordings. Customers would leverage the Mattersight portal to access call recordings and meta data as no direct access to the file structure is provided.

The detailed administration of basic connectivity between Communication Manager, Application Enablement Services, and contact center devices is not the focus of these Application Notes and will not be described.

In the compliance testing, the contact center devices consisted of two VDNs, two skill groups, one supervisor, and two agents stations as shown in the table below.

Device Type	Extension		
VDN	56001, 56010		
Skill/Hunt Group	56300, 56303		
Agent ID	1000, 1004		
Agent Station	56201, 56101		
Supervisor	56202		

A new hunt group will need to be created so the Mattersight CRS application can keep track of the agents that need to be recorded in this case skill group 5 with extension is 56304 was used. All agents that are in-scope for call recording would need to have this skill assigned to them in this example agent 1000 and 1004 were added to skill group 5.

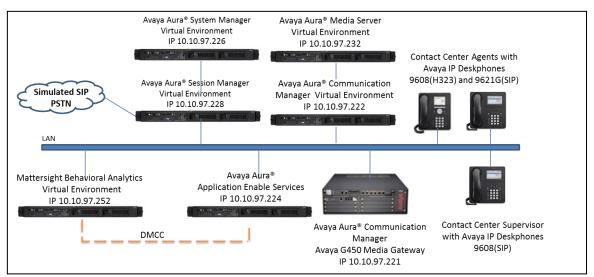


Figure 1: Compliance Testing Configuration

4. Equipment and Software Validated

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

Equipment/Software	Release/Version		
Avaya Aura® Communication Manager on Virtual Environment	7.0.2		
Avaya G450 Media Gateway	7.0.1.1		
Avaya Aura® Media Server on Virtual Environment	7.7.0.334		
Avaya Aura® Application Enablement Services	7.0.1 Super Patch 3		
Avaya 9608, 9621G IP Deskphone (H.323)	6.6.3		
Avaya 9621G IP Deskphone (SIP)	7.0.1.4		
Avaya Aura® Session Manager	7.0.1 SP2		
Avaya Aura® System Manager	7.0.1.2		
MatterSight Call Recording Solution on Windows 2012R2 Server • Avaya DMCC .NET (ServiceProvider.dll)	4.0 SP 2 6.2.0.29		

5. Configure Avaya Aura® Communication Manager

This section provides the procedures for configuring Communication Manager. The procedures include the following areas:

- Verify License
- Administer CTI Link
- Administer System Parameters Features
- Administer Recording Hunt Group
- Administer Virtual IP Softphones

5.1. Verify License

Log in to the System Access Terminal (SAT) to verify that the Communication Manager license has proper permissions for features illustrated in these Application Notes. Use the "display system-parameters customer-options" command to verify that the **Computer Telephony Adjunct Links** customer option is set to "y" on **Page 4**. If this option is not set to "y", then contact the Avaya sales team or business partner for a proper license file.

```
display system-parameters customer-options
                                                                    4 of 12
                                                              Page
                               OPTIONAL FEATURES
       Access Security Gateway (ASG)? n

Analog Trunk Incoming Call ID? v
   Abbreviated Dialing Enhanced List? y
A/D Grp/Sys List Dialing Start at 01? y
                                                                CAS Main? n
                                                      Change COR by FAC? n
Answer Supervision by Call Classifier? y
                                ARS? y Computer Telephony Adjunct Links? y
                ARS/AAR Partitioning? y Cvg Of Calls Redirected Off-net? y
         ARS/AAR Dialing without FAC? n
                                                            DCS (Basic)? y
         ASAI Link Core Capabilities? y
                                                      DCS Call Coverage? y
         ASAI Link Plus Capabilities? y
                                                      DCS with Rerouting? y
      Async. Transfer Mode (ATM) PNC? n
                                         Digital Loss Plan Modification? y
 Async. Transfer Mode (ATM) Trunking? n
             ATM WAN Spare Processor? n
                                                                DS1 MSP? y
                               ATMS? y
                                                  DS1 Echo Cancellation? y
                 Attendant Vectoring? y
        (NOTE: You must logoff & login to effect the permission changes.)
```

5.2. Administer CTI Link

Add a CTI link using the "add cti-link n" command, where "n" is an available CTI link number. Enter an available extension number in the **Extension** field. Note that the CTI link number and extension number may vary. Enter "ADJ-IP" in the **Type** field, and a descriptive name in the **Name** field. Default values may be used in the remaining fields.

```
add cti-link 1

CTI Link

CTI Link: 1

Extension: 56001

Type: ADJ-IP

COR: 1

Name: DevvmAES
```

5.3. Administer System Parameters Features

Use the "change system-parameters features" command to enable **Create Universal Call ID** (**UCID**), which is located on **Page 5**. For **UCID Network Node ID**, enter an available node ID.

```
5 of 19
change system-parameters features
                                                             Page
                       FEATURE-RELATED SYSTEM PARAMETERS
SYSTEM PRINTER PARAMETERS
 Endpoint:
                        Lines Per Page: 60
SYSTEM-WIDE PARAMETERS
                                    Switch Name:
           Emergency Extension Forwarding (min): 10
         Enable Inter-Gateway Alternate Routing? n
Enable Dial Plan Transparency in Survivable Mode? n
                            COR to Use for DPT: station
               EC500 Routing in Survivable Mode: dpt-then-ec500
MALICIOUS CALL TRACE PARAMETERS
              Apply MCT Warning Tone? n MCT Voice Recorder Trunk Group:
     Delay Sending RELease (seconds): 0
SEND ALL CALLS OPTIONS
    Send All Calls Applies to: station Auto Inspect on Send All Calls? n
             Preserve previous AUX Work button states after deactivation? n
UNIVERSAL CALL ID
    Create Universal Call ID (UCID)? y
                                         UCID Network Node ID: 1
```

Navigate to **Page 13**, and enable **Send UCID to ASAI**. This parameter allows for the universal call ID to be sent to the Call Recording Solution.

```
change system-parameters features
                                                               Page 13 of 19
                        FEATURE-RELATED SYSTEM PARAMETERS
CALL CENTER MISCELLANEOUS
          Callr-info Display Timer (sec): 10
                        Clear Callr-info: next-call
       Allow Ringer-off with Auto-Answer? n
   Reporting for PC Non-Predictive Calls? n
           Agent/Caller Disconnect Tones? n
         Interruptible Aux Notification Timer (sec): 3
            Zip Tone Burst for Callmaster Endpoints: double
 ASAI
           Copy ASAI UUI During Conference/Transfer? n
       Call Classification After Answer Supervision? n
                                  Send UCID to ASAI? y
         For ASAI Send DTMF Tone to Call Originator? y
 Send Connect Event to ASAI For Announcement Answer? n
```

5.4. Administer Hunt Group

A new hunt group will need to be created so the Mattersight CRS application can keep track of the agents that need to be recorded. All agents that are in-scope for call recording would need to have this skill assigned to them. Below is the configuration for **hunt group** 5.

```
1 of
display hunt-group 5
                                                                Page
                                 HUNT GROUP
           Group Number: 5
                                                           ACD? y
             Group Name: Basic 3
                                                         Queue? y
         Group Extension: 56304
                                                        Vector? y
             Group Type: ucd-mia
                     TN: 1
                    COR: 1
                                              MM Early Answer? n
          Security Code:
                                      Local Agent Preference? n
 ISDN/SIP Caller Display:
           Queue Limit: unlimited
 Calls Warning Threshold:
                             Port:
 Time Warning Threshold:
                              Port:
```

5.5. Administer Virtual IP Softphones

Virtual IP Softphones are used by Mattersight to single step conference with the agent phone and capture media. Add a virtual IP softphone using the "add station n" command, where "n" is an available extension number. Enter the following values for the specified fields, and retain the default values for the remaining fields.

• **Extension:** The available extension number.

• **Type:** Any IP telephone type, such as "9620".

• Name: A descriptive name.

• Security Code: A desired code, example: 1234. Below screenshot, security code has been

mask by Communication Manager for security reason.

• **IP SoftPhone:** Set to "y".

```
add station 56114
                                                                               5
                                                                Page
                                                                       1 of
                                     STATION
                                     Lock Messages? n
Security Code: *-
Coverage Path 1:
Coverage Path 2:
Extension: 56114
                                                                       BCC: 0
    Type: 9620
                                                                        TN: 1
                                                                       COR: 1
    Port: S00067
                                                                       cos: 1
    Name: BA Virtual #1
                                      Hunt-to Station:
                                                                     Tests: y
STATION OPTIONS
                                           Time of Day Lock Table:
             Loss Group: 19 Personalized Ringing Pattern: 1
       Speakerphone: 2-way
Display Language: english
                                                Message Lamp Ext: 56114
                                            Mute Button Enabled? y
                                                Expansion Module? n
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
                                              Customizable Labels? Y
```

Repeat this section to administer the desired number of virtual IP softphones, using sequential extension numbers. In the compliance testing, two virtual IP softphones were administered as shown below, to allow for simultaneous recording of two monitored agents in **Section 3**.

list station 56114 count 2							
STATIONS							
Ext/ Hunt-to	Port/ Type	Name/ Surv GK NN	Move	Room/ Data Ext	Cv1/ COR/ Cable/ Cv2 COS TN Jack		
56114 56115	\$00067 9620 \$00070	BA Virtual #1	no		1 1 1		
	9620	" _	no		1		

6. Configure Avaya Aura® Application Enablement Services

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

- Launch OAM interface
- Verify license
- Administer TSAPI link
- Administer H.323 gatekeeper
- Disable security database
- Restart services
- Administer Call Recording Solution user
- Provide Call Recording Solution User unrestricted access
- Administer ports

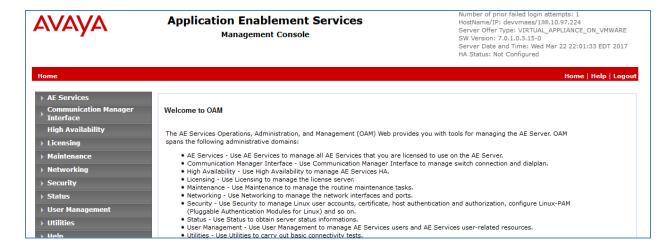
6.1. Launch OAM Interface

Access the OAM web-based interface by using the URL "https://ip-address" in an Internet browser window, where "ip-address" is the IP address of the Application Enablement Services server.

The **Please login here** screen is displayed. Log in using the appropriate credentials.

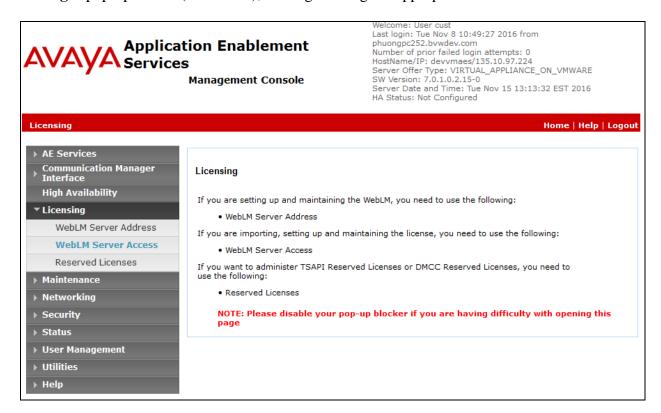


The **Welcome to OAM** screen is displayed as below.



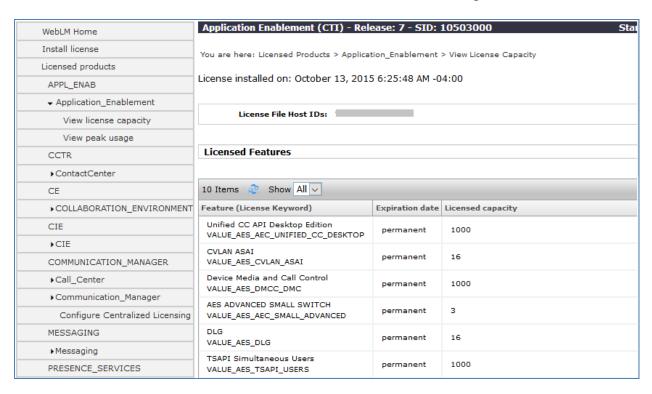
6.2. Verify License

Select Licensing \rightarrow WebLM Server Access in the left pane, to display the Web License Manager pop-up screen (not shown), and log in using the appropriate credentials.



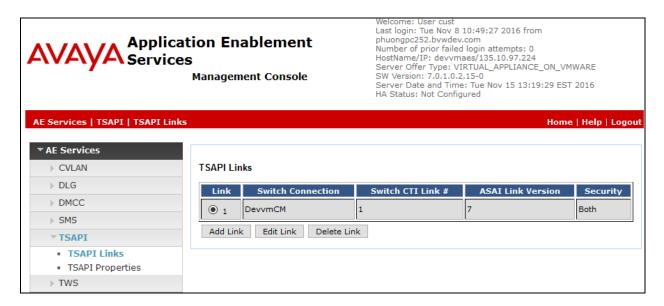
The WebLM screen below is displayed. Select Licensed products → APPL_ENAB → Application_Enablement in the left pane, to display the Application Enablement (CTI) screen in the right pane.

Verify that there are sufficient licenses for **TSAPI Simultaneous Users** and **Device Media and Call Control**, as shown below. Note that the TSAPI license is used for monitoring and call control via DMCC, and the DMCC license is used for the virtual IP softphones.



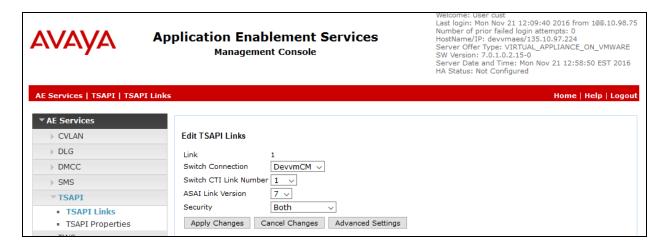
6.3. Administer TSAPI Link

To administer a TSAPI link, select **AE Services** → **TSAPI** → **TSAPI Links** from the left pane. The **TSAPI Links** screen is displayed, as shown below. Click **Add Link**.



The **Add TSAPI Links** screen is displayed next (not shown), below is example of link created during compliance test.

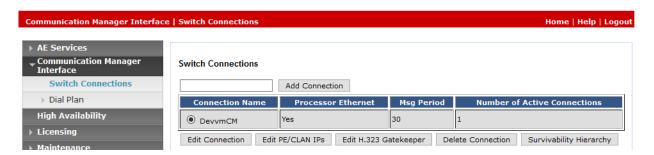
The **Link** field is only local to the Application Enablement Services server, and may be set to any available number. For **Switch Connection**, select the relevant switch connection from the drop-down list. In this case, the existing switch connection "DevvmCM" is selected. For **Switch CTI Link Number**, select the CTI link number from **Section 5.2**. Select "7" for **ASAI Link Version** and select "Both" for **Security.**



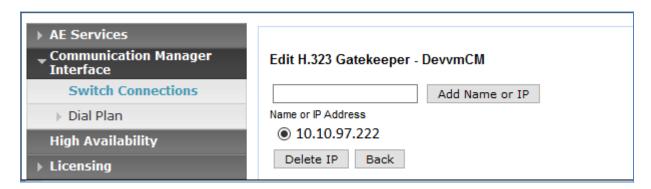
6.4. Administer H.323 Gatekeeper

Select Communication Manager Interface \rightarrow Switch Connections from the left pane. The Switch Connections screen shows a listing of the existing switch connections.

Locate the connection name associated with the relevant Communication Manager, in this case "S8300D", and select the corresponding radio button. Click **Edit H.323 Gatekeeper**.

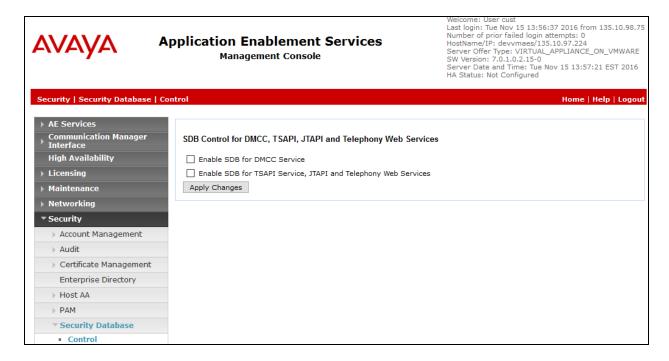


The **Edit H.323 Gatekeeper** screen is displayed. Enter the IP address of a C-LAN circuit pack or the Processor C-LAN on Communication Manager to be used as H.323 gatekeeper, click **Add Name or IP**. Example below is **H323 Gatekeeper** already created with IP address of Communication Manager 10.10.97.222



6.5. Disable Security Database

Select Security \rightarrow Security Database \rightarrow Control from the left pane, to display the SDB Control for DMCC, TSAPI, JTAPI and Telephony Web Services screen in the right pane. Uncheck both fields below, and click Apply Changes.



6.6. Restart Services

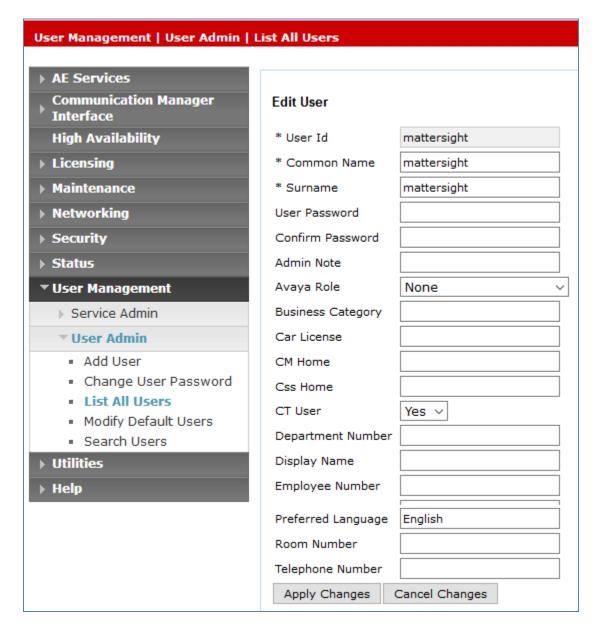
Select Maintenance \rightarrow Service Controller from the left pane, to display the Service Controller screen in the right pane. Check DMCC Service and TSAPI Service, and click Restart Service.



6.7. Administer Call Recording Solution User

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. Below is screenshot of user mattersight created during compliance tested.



6.8. Provide Call Recording Solution User unrestricted access

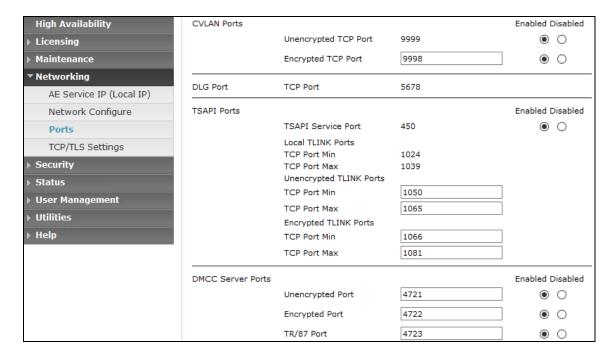
Select Security \rightarrow Security Database \rightarrow CTI Users \rightarrow List All Users from the left pane to display the list of CTI users in the right pane (not shown). Under the user list select the user created in the previous step and click on Edit at the bottom of the page. In the User Profile section check the box for the Unrestricted Access.

This will allow the CTI user access to all devices on this AES server.

6.9. Administer Ports

Select **Networking \rightarrow Ports** from the left pane, to display the **Ports** screen in the right pane.

In the **DMCC Server Ports** sub-section, select the radio button for **Unencrypted Port** under the **Enabled** column, and make a note of the port value to be used later to configure Call Recording Solution. Retain the default values in the remaining fields. Click **Apply Changes** at the bottom of the screen (not shown below).



7. Configure Mattersight Call Recording Solution

This section provides the procedures for configuring Call Recording Solution. The procedures include the configuration of the WorkerSettings.config file.

The configuration of Call Recording Solution is performed by Mattersight technicians. The procedural steps are presented in these Application Notes for informational purposes.

7.1. Administer WorkerSettings

In the WorkerSettings file configure the following parameters in order for Mattersight to communicate with Application Enablement Service:

- **RecordingExtensions**: Enter virtual extension created in **Section 5.5**.
- **CmIpAddresses**: Enter IP address of Communication Manager.
- **RtpIpAddress**: Enter IP address of Mattersight server
- **AesIpAddress**: Enter IP address of Application Enablement Service.
- AesSoketPort: Enter port in Section 6.9.
- **DmccUserName**: Enter user name create in **Section 6.7**.
- **DmccPassword**: Enter password created in **Section 6.7**.
- Codec: Enter codec used for recording
- **Mode**: Enter the interface mode in this case it is SSC.

```
<workerSettings>
 <add key="EndpointsToPublishTo" value="tcp://127.0.0.1:56000" />
 <add key="EndpointsToSubscribeTo" value="tcp://127.0.0.1:56001" />
 <add key="ExtensionsWhitelist" value="" />
 <add key="RecordingExtensions" value="56114-56115" />
 <add key="CmIpAddress" value="100.10.97.222" />
 <add key="StartPort" value="4800" />
 <add key="RtpIpAddresses" value="1" .10.97.252" />
 <add key="AesIpAddress" value="1: 10.97.224" />
 <add key="AesSocketPort" value="4721" />
 <add key="DmccUserName" value="mattersight" />
 <add key="DmccPassword" value="mattersight" />
 <add key="ProtocolVersion" value="http://www.ecma-international.org/standards/ecma-323/csta/ed3/priv6" />
 <add key="RecordOnlyAgentToAgent" value="false" />
 <add key="Codec" value="g711U" /><!--g711U, g729, g711A-->
 <add key="Mode" value="SSC" /><!--MR (multi registration), SSC (single step conf), SO (Service Observe)--</pre>
 <add key="StationPassword" value="1234" />
 <add key="FAC" value="*43" /><!--feature access code for service observe-->
 <add key="ServerPingIntervalInSecs" value="10" />
 <add key="ReconnectDelayInMins" value="2" />
 /workerSettings>
```

Close the file and save all of the configuration changes.

7.2. Administer Service

In Services window, verify Mattersight Dmcc Avaya Interface is running as shown below (this is performed by the Mattersight team):



8. Verification Steps

This section provides the tests that can be performed to verify proper configuration of Communication Manager, Application Enablement Services, and Call Recording Solution.

8.1. Verify Avaya Aura® Communication Manager

On Communication Manager, verify the status of the administered CTI link by using the "status aesvcs cti-link" command. Verify that the **Service State** is "established" for the CTI link number administered in **Section 5.2**, as shown below.

statu	status aesvcs cti-link								
AE SERVICES CTI LINK STATUS									
CTI Link	Version	Mnt Busy	AE Services Server		Service State	Msgs Sent	Msgs Rcvd		
1	7	no	devaes	est	ablished	26	20		

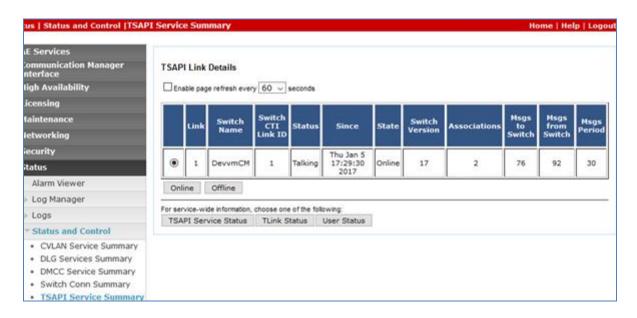
Verify the registration status of the virtual softphones by using the "list registered-ip-stations" command. Verify that all virtual extensions from **Section 5.5** are displayed, as shown below.

```
list registered-ip-stations
                                                                                    Page
                                 REGISTERED IP STATIONS
Station Ext Set Type/ Prod ID/ Station IP Address/ or Orig Port Net Rgn Release Skt Gatekeeper IP Address
56101
                9608 IP_Phone tcp 10.33.5.29
                            6.6229 10.10.97.2222
IP_API_A top 10.10.97.224
                9620
56114
                            3.2040
                                          10.10.97.222
                9620
                            IP_API_A tcp 10.10.97.224
56115
                            3.2040
                                             10.10.97.222
```

8.2. Verify Avaya Aura® Application Enablement Services

On Application Enablement Services, verify the status of the TSAPI link by selecting **Status Status and Control TSAPI Service Summary** from the left pane. The **TSAPI Link Details** screen is displayed.

Verify the **Status** is "Talking" for the TSAPI link administered in **Section 6.3**, and that the **Associations** column reflects the number of monitored skill group and agent station extensions from **Section 3**.



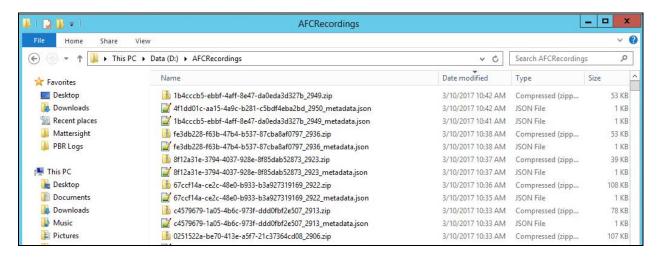
Verify the status of the DMCC link by selecting **Status** → **Status** and **Control** → **DMCC Service Summary** from the left pane. The **DMCC Service Summary** – **Session Summary** screen is displayed.

Verify the **User** column shows an active session with the Call Recording Solution user name from **Section 6.7**, and that the # **of Associated Devices** column reflects the number of monitored skill group, agent station extensions, and virtual IP softphone extensions.



8.3. Verify Mattersight Call Recording Solution

In the AFCRecordings folder, select the entry and verify that the call recording can be played back and play the correct media.



9. Conclusion

These Application Notes describe the configuration steps required for Mattersight Call Recording Solution 4.0 to successfully interoperate with Avaya Aura® Communication Manager 7.0.2 using Avaya Aura® Application Enablement Services 7.0.1 SP3. All feature and serviceability test cases were completed successfully with observations noted in **Section 2.2**.

10. Additional References

This section references the product documentation relevant to these Application Notes can be found on support.avaya.com:

- **1.** Administering Avaya Aura® Communication Manager, Release 7.0.1 555-245-205 Issue 3 October 2016.
- **2.** Avaya Aura® Application Enablement Services Administration and Maintenance Guide, Release 7.0.1 Issue 2 August 2016.

Mattersight document is available upon request.

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