

Avaya Solution & Interoperability Test Lab

Application Notes for Grandsys LOG8000 with Avaya AuraTM Communication Manager and Avaya AuraTM Application Enablement Services - Issue 1.0

Abstract

These Application Notes describe the procedures for configuring Grandsys LOG8000 to monitor and record calls placed to and from Avaya IP telephones and agents on Avaya AuraTM Communication Manager. Grandsys LOG8000 uses the Device, Media and Call Control (DMCC) API of the Avaya AuraTM Application Enablement Services to monitor stations to obtain call information and to register DMCC softphones that Grandsys LOG8000 uses as recording ports.

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 a compliance-tested configuration comprised of an Avaya AuraTM Communication Manager, an Avaya AuraTM Application Enablement Services and Grandsys LOG8000.

Grandsys LOG8000 is a recording solution made for the customers of the call center market. Grandsys LOG8000 communicates with Application Enablement Services using the Device, Media and Call Control (DMCC) API to monitor stations to obtain call information and to register DMCC softphones that Grandsys LOG8000 uses as recording ports. When a call starts on an extension to be recorded, Communication Manager will send the audio stream to Grandsys LOG8000 which will then record the call and save the recording to the database. Detailed call information obtained using DMCC is also stored for each call along with the recording.

1.1. Interoperability Compliance Testing

The interoperability compliance test included feature and serviceability testing. The feature testing evaluated the ability of Grandsys LOG8000 to monitor and record calls placed to and from stations and agents. The serviceability testing introduced failure scenarios to see if Grandsys LOG8000 can resume recording after failure recovery.

1.2. Support

For technical support on Grandsys LOG8000, contact Grandsys at:

Phone: +886-2-87682715Email: service@grandsys.com

2. Reference Configuration

Figure 1 illustrates a sample configuration consisting of an Avaya S8500 Server, an Avaya G650 Media Gateway, an Application Enablement Services Server, Avaya IP Telephones and a Windows 2003 Server running Grandsys LOG8000. The Grandsys LOG8000 Server monitors the agent extensions using the DMCC Service to record the call and obtain call related information. The DMCC Service is provided by the Application Enablement Services Server.

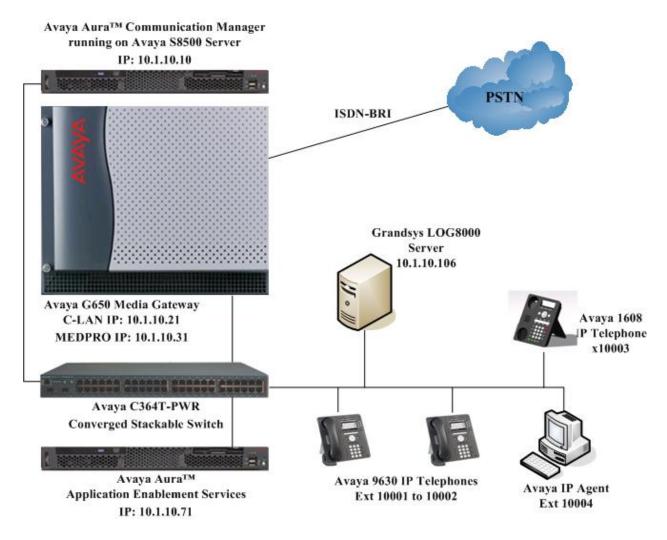


Figure 1: Test Configuration

3. Equipment and Software Validated

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

Equipment	Version
Avaya S8500 Server	Communication Manager
	5.2
	(Service Pack 02.0.947.3-17250)
Avaya G650 Media Gateway	-
- TN2312BP IP Server Interface	HW07, FW046
- TN799DP C-LAN Interface	HW01, FW032
- TN2602AP IP Media Processor	HW02, FW048
- TN2464BP DS1 Interface	HW05, FW022
Application Enablement Services	4.2.1 with Patch 2
Avaya 9630 IP Telephones	3.0 (H.323)
Avaya 1608 IP Telephone	1.1.0.0 (H.323)
Avaya IP Agent	7.0.32.198
Avaya C364T-PWR Converged Stackable Switch	4.5.18
Grandsys LOG8000	2.2.2

4. Configure Communication Manager

This section provides the procedures for configuring an ip-codec-set and ip-network region, a switch connection and Computer Telephony Integration (CTI) links and recorded/monitored stations on Communication Manager. All the configuration changes in Communication Manager are performed through the System Access Terminal (SAT) interface. The highlights in the following screens indicate the values used during the compliance test.

4.1. Codec Configuration

Enter the **change ip-codec-set n** command, where **n** is a number between 1 and 7, inclusive. Ensure that the supported codecs for DMCC are used. In this compliance testing, the **G.711MU** codec and no media encryption are used.

chai	nge ip-codec-	set 1			Page	1 of	2
	Codec Set: 1		Codec Set				
	Audio Codec	Silence Suppression	Frames Per Pkt	Packet Size(ms)			
1:	G.711MU	n	2	20			
2:	G.711A	n	2	20			
3:	G.729	n	2	20			
1:	Media Encry	ption					

4.2. IP Network Regions

During compliance testing, a C-LAN board dedicated for H.323 endpoint registration was assigned to IP network region 1. The Avaya IP Telephones and IP Softphones used by Grandsys LOG8000 registered with the C-LAN board and were thus also assigned to IP network region 1. As a result, the RTP traffic between them is governed by the codec set defined by the Codec Set field. In this configuration, IP codec set 1 is used as defined in Section 4.1.

```
change ip-network-region 1
                                                                        Page
                                                                                1 of 19
                                   IP NETWORK REGION
  Region: 1
                  Authoritative Domain:
Location:
   Name:
MEDIA PARAMETERS
                                   Intra-region IP-IP Direct Audio: yes
      Codec Set: 1
                                  Inter-region IP-IP Direct Audio: yes
   UDP Port Min: 2048
                                                IP Audio Hairpinning? n
   UDP Port Max: 3929
DIFFSERV/TOS PARAMETERS
DIFFSERV/TOS PARAMETERS RTCP Reporting Enabled? y
Call Control PHB Value: 46 RTCP MONITOR SERVER PARAMETERS
Audio PHB Value: 46 Use Default Server Parameters? y
                                             RTCP Reporting Enabled? y
        Video PHB Value: 46
802.1P/Q PARAMETERS
Call Control 802.1p Priority: 6
        Audio 802.1p Priority: 6
        Video 802.1p Priority: 5
                                          AUDIO RESOURCE RESERVATION PARAMETERS
H.323 IP ENDPOINTS
                                                                RSVP Enabled? n
  H.323 Link Bounce Recovery? y
 Idle Traffic Interval (sec): 20
   Keep-Alive Interval (sec): 5
             Keep-Alive Count: 5
```

4.3. Configure AES and CTI Links

Grandsys LOG8000 uses the DMCC API to communicate with Application Enablement Services and to obtain call information, which requires the Telephony Services Application Programming Interface (TSAPI) CTI link to be configured on Communication Manager. Application Enablement Services communicates with Communication Manager over an AES link. Within the AES link, a CTI link is configured to provide the required TSAPI service to Grandsys LOG8000. The following steps demonstrate the configuration of the Communication Manager side of the AES and CTI links. See **Section 5** for the details of configuring the Application Enablement Services side of the AES and TSAPI CTI links.

Step	Description
1.	Enter the display system-parameters customer-options command. On Page 3, verify that Computer Telephony Adjunct Links is set to y . If not, contact an authorized Avaya account representative to obtain the license.
	display system-parameters customer-options Page 3 of 11 OPTIONAL FEATURES
	Abbreviated Dialing Enhanced List? n Access Security Gateway (ASG)? n Analog Trunk Incoming Call ID? n A/D Grp/Sys List Dialing Start at 01? n Answer Supervision by Call Classifier? n ARS? y ARS/AAR Partitioning? y ARS/AAR Dialing without FAC? n ASAI Link Core Capabilities? n ASAI Link Plus Capabilities? n Async. Transfer Mode (ATM) PNC? n ATM WAN Spare Processor? n Attendant Vectoring? n Access Security Gateway (ASG)? n Authorization Codes? y Authorization Codes? y Authorization Codes? y Authorization Codes? y CAS Branch? n CAS Main? n Computer Telephony Adjunct Links? y Cvg Of Calls Redirected Off-net? n DCS (Basic)? n DCS Call Coverage? n DCS with Rerouting? n Digital Loss Plan Modification? n DS1 MSP? n DS1 Echo Cancellation? n
2.	Enter the add cti-link n command, where n is a number between 1 and 64, inclusive. Enter a valid Extension under the provisioned dial plan in Communication Manager, set the Type field to ADJ-IP, and assign a descriptive Name to the CTI link. add cti-link 1 CTI Link: 1 Extension: 19951
	Type: ADJ-IP COR: 1 Name: TSAPI Svcs
3.	Enter the change node-names ip command. In the compliance-tested configuration, the CLAN-01A02 IP address was utilized for registering H.323 endpoints (Avaya IP Telephones) and for connectivity to Application Enablement Services.
	Change node-names ip IP NODE NAMES Name IP Address CLAN-01A02 MEDPRO-01A13 VAL-01A04 default 0.0.0.0 procr 10.1.10.10

Description Step 4. Enter the **change ip-services** command. On Page 1, configure the **Service Type** field to **AESVCS** and the **Enabled** field to v. The **Local Node** field should be pointed to the CLAN-01A02 board that was configured previously in Step 3. During the compliance test, the default port was utilized for the **Local Port** field. change ip-services 1 of Page IP SERVICES Service Enabled Local Local Remote Remote Node Node Type Port Port CLAN-01A02 AESVCS 8765 On Page 3, enter the hostname of the Application Enablement Services server for the AE **Services Server** field. The server name may be obtained by logging in to the Application Enablement Services server using Secure Shell (SSH), and running the uname -a command. Enter an alphanumeric password for the **Password** field and set the **Enabled** field to y. The same password will be configured on the Application Enablement Services server in **Section 5.1**. change ip-services 3 of Page AE Services Administration Server ID AE Services Enabled Password Status Server 1: xxxxxxxxxxxx aes1 У 2: 3:

4.4. Recorded (Monitored) Stations

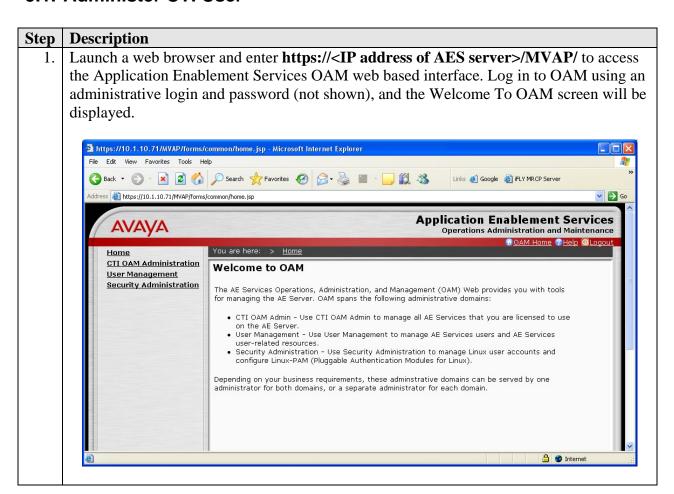
Step	Description					
1.	Enter the change station n correquired to be recorded. On P to y and specify the Security 10001 to 10004 were modified	age 1 of the Code. For t	STATION form, set the IP S	oftpho	ne field	
	change station 10001			Page	1 of	5
			STATION	1 4 9 0	1 01	
	Extension: 10001 Type: 9630 Port: S00002 Name: Alice		Lock Messages? n Security Code: 000000 Coverage Path 1: Coverage Path 2:		BCC: TN: COR: COS:	1
	STATION OPTIONS		Hunt-to Station:			
	Loss Group: 1	19	Time of Day Lock Tab Personalized Ringing Patte Message Lamp E	rn: 1	001	
	Speakerphone: 2 Display Language: 6 Survivable GK Node Name:	-	Mute Button Enabl Button Modul	-		
	Survivable COR: i		Media Complex E IP SoftPho			
			IP Video Softpho	ne? n		
			Customizable Labe	ls? y		

5. Configure Application Enablement Services

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

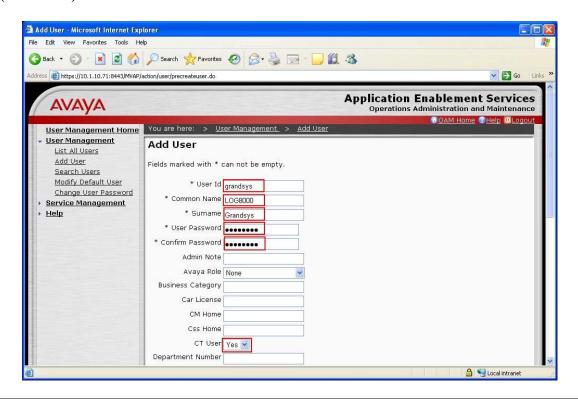
- Administer CTI User
- Verify Application Enablement Services License
- Administer Switch Connection
- Administer TSAPI link
- Administer Ports
- Administer CTI user permission

5.1. Administer CTI User

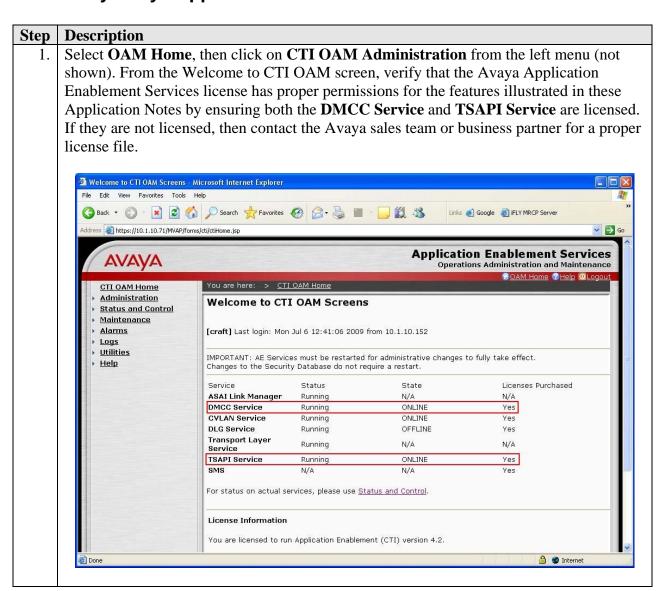


Step | **Description**

2. Click **User Management**, then **User Management** > **Add User** in the left pane. Specify a value for **User Id**, **Common Name**, **Surname**, **User Password** and **Confirm Password**. Set **CT User** to **Yes**. Use the values for **User Id** and **User Password** to configure Grandsys LOG8000 in **Section 6.1 Step 1** to access the DMCC Services on the Application Enablement Services. Scroll down to the bottom of the page and click **Apply** (not shown).



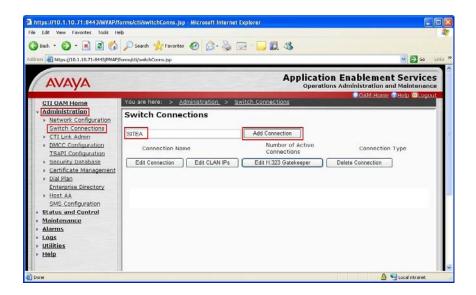
5.2. Verify Avaya Application Enablement Services License



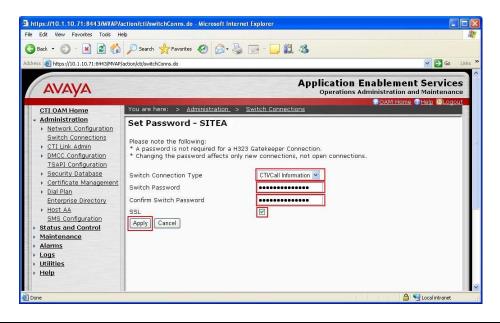
5.3. Administer Switch Connection

SITEA is used.

Step Description From the CTI OAM Home menu, select Administration > Switch Connections. Enter a descriptive name for the switch connection and click Add Connection. In this case,

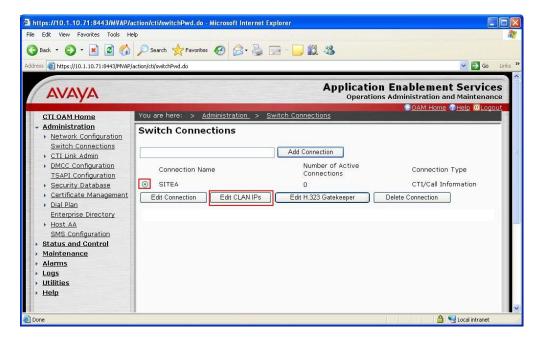


2. The Set Password screen is displayed. Select **CTI/Call Information** for **Switch Connection Type**. For the **Switch Password** and **Confirm Switch Password** fields, enter the password that was administered in Communication Manager using the IP Services form in **Section 4.3 Step 4**. The **SSL** field needs to be checked for the S8500 Server. Click **Apply**.

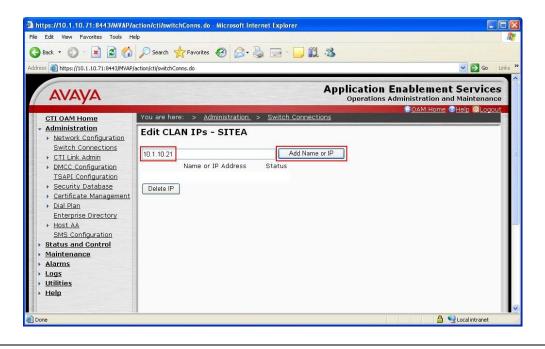


Step | **Description**

3. The Switch Connections screen is displayed. Select the newly added switch connection name and click **Edit CLAN IPs**.

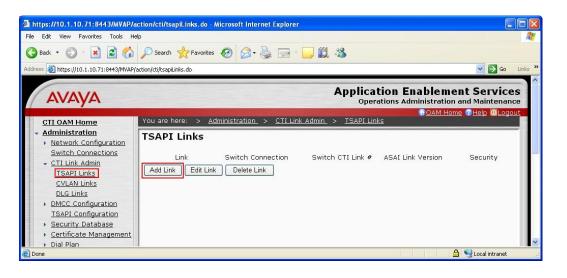


4. In the Edit CLAN IPs screen, enter the host name or IP address of the C-LAN used for Application Enablement Services connectivity. In this case, **10.1.10.21** is used, which corresponds to the IP address of the C-LAN administered on Communication Manager in **Section 4.3 Step 3**. Click **Add Name or IP**.



5.4. Administer TSAPI Link

Step Description
 To administer a TSAPI link on AES, select Administration > CTI Link Admin > TSAPI Links from the CTI OAM Home menu. Click Add Link.



2. In the Add / Edit TSAPI Links screen, select the following values:

• Link: Select an available Link number from 1 to 16.

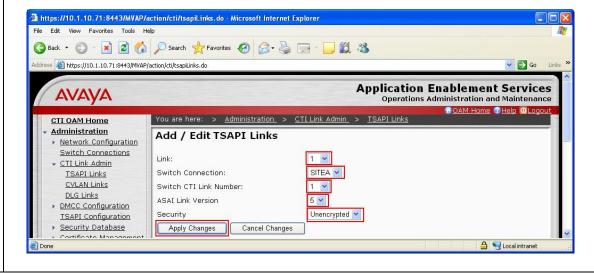
• **Switch Connection:** Administered switch connection in **Section 5.3 Step 1**.

• Switch CTI Link Number: Corresponding CTI link number in Section 4.3 Step 2.

• **ASAI Link Version:** Set to 5.

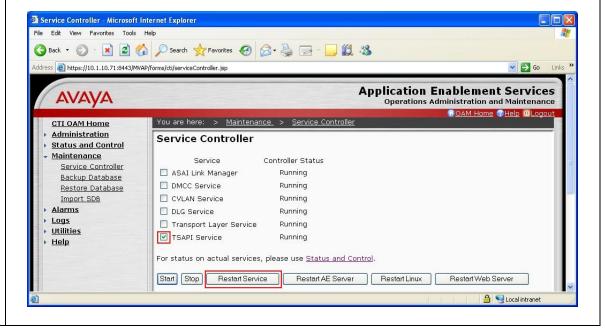
• **Security: Unencrypted** TSAPI Links are used.

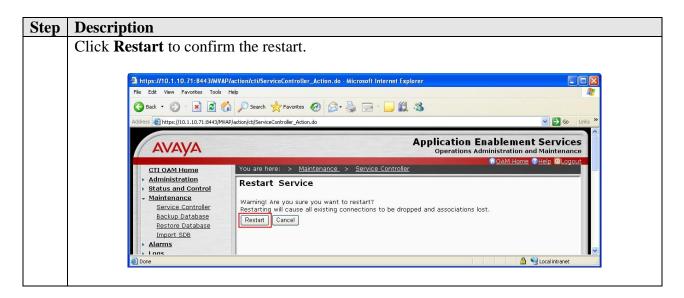
Note that the actual values may vary. Click **Apply Changes**.





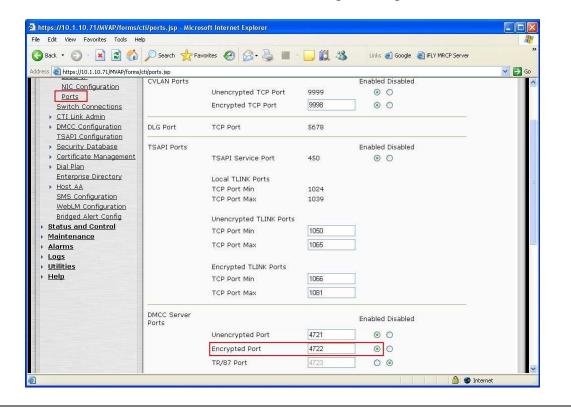
3. To restart the TSAPI Service, select **Maintenance > Service Controller** from the CTI OAM Home menu. Check the **TSAPI Service** checkbox and click **Restart Service**.





5.5. Administer Ports

Navigate to the CTI OAM Home > Administration > Network Configuration > Ports page to set the DMCC Server Ports. During the compliance test, the default port values were utilized as shown below. Since the encrypted port was utilized during the compliance test, set the Encrypted Port field to Enabled. Click the Apply Changes button (not shown) at the bottom of the screen to complete the process.

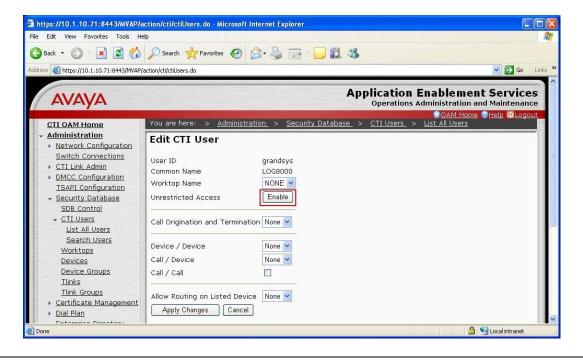


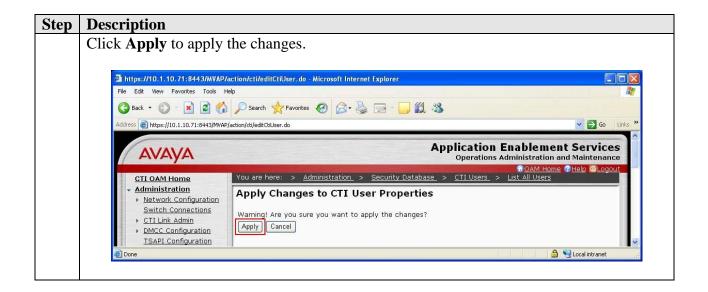
5.6. Administer CTI User Permission

Step Description
 Select Administration > Security Database > CTI Users > List All Users from the CTI OAM Home menu. Select the User ID created in Section 5.1 Step 2 and click Edit.



2. Assign access rights and call/device privileges according to customer requirements. For simplicity in configuration, **Unrestricted Access** was enabled during compliance testing. If **Unrestricted Access** is not desired, then consult [1] for guidance on configuring the call/device privileges as well as devices and device groups. Click **Enable**.



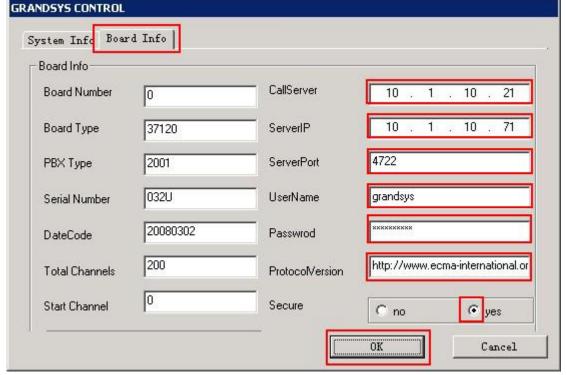


6. Configure Grandsys LOG8000

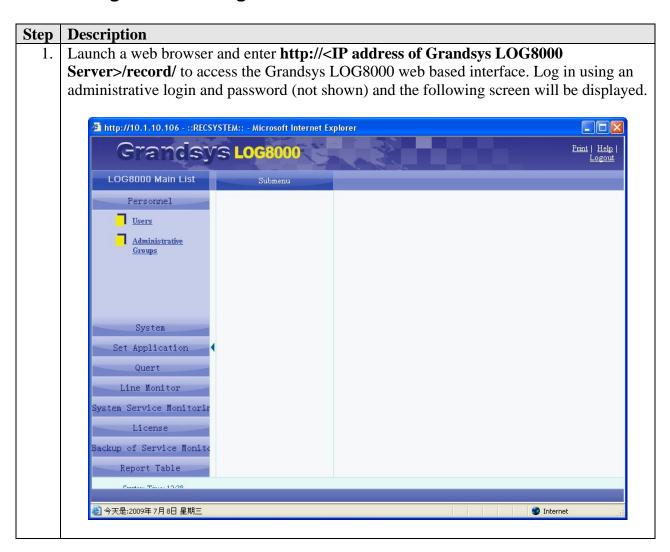
Grandsys installs, configures, and customizes the Grandsys LOG8000 application for their end customers. This section only describes the interface configuration for the Grandsys LOG8000 application to communicate with Application Enablement Services and Communication Manager. Refer to [3] and [4] for configuring the Grandsys LOG8000 application.

6.1. Configure DMCC Settings

1. On the Grandsys LOG8000 Server, execute C:\Program Files\Grandsys Software\Record System\Record Service\Grandsys_Contral.exe to configure the DMCC settings. Click on the Board Info tab and configure the following: • CallServer: Set to the C-LAN IP address in Section 4.3 Step 3 for the registration of the DMCC stations. • ServerIP: Set to the IP address of the Application Enablement Services. • ServerPort: Set to the encrypted DMCC Server Port in Section 5.5. • UserName: Set to the User Id field in Section 5.1 Step 2. • Password: Set to the User Password field in Section 5.1 Step 2. • ProtocolVersion: Set to the value http://www.ecmainternational.org/standards/ecma-323/csta/ed3/priv3. • Secure: Set to yes since encryption is used.



6.2. Configure Recording Stations

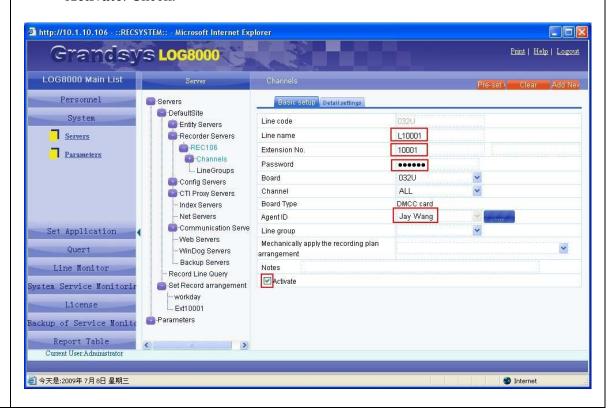


Step | **Description**

2. Click **System > Servers** on the left-most pane. To configure the recording stations, expand **Servers > DefaultSite > Recorder Servers > REC106** in the center pane and click on **Channels**. Note: "REC106" is the name defined for this compliance test and will vary.

On the right-most pane, enter the following values in the **Basic Setup** tab:

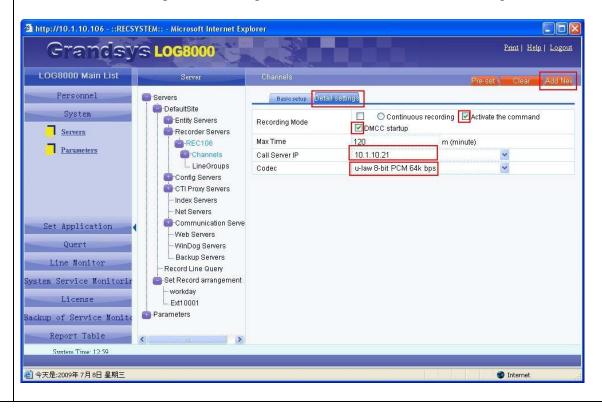
- **Line Name**: Enter a descriptive name.
- Extension No.: Phone extension to be recorded.
- Password: Enter the Phone Security Code configured in Section 4.4, Step 1.
- **Agent ID**: Select from the list a user configured in Grandsys LOG8000.
- Activate: Check.



Step | Description

- 3. Click the **Detail Settings** tab and configure the following:
 - Activate the command: Check.
 - **DMCC Startup**: Check.
 - Call Server IP: Select the IP Address of the C-LAN configured in Section 6.1 Step 1.
 - Codec: Select the codec to match the one configured on Communication Manager in Section 4.1.

Click **Add New** to save the settings. Repeat this step for all the agent stations to be recorded. In this configuration, the agent stations 10001 to 10004 are configured.



7. General Test Approach and Test Results

The general approach was to place various types of calls to and from stations, agents, and Vector Directory Numbers (VDNs), monitor and record the calls using Grandsys LOG8000, and verify the recordings. For feature testing, the types of calls included internal calls, inbound and outbound trunk calls, transferred calls, and conferenced calls. For serviceability testing, failures such as disconnecting the LAN cable to the Grandsys LOG8000 Server and Application Enablement Services, and resetting the Grandsys LOG8000 Server and Communication Manager were applied.

All test cases were executed and passed.

8. Verification Steps

This section provides the tests that can be performed to verify proper configuration of Communication Manager, Application Enablement Services and Grandsys LOG8000.

8.1. Verify Communication Manager

Verify the status of the administered AE Services Link by using the **status aesvcs link** command. The following shows that the link between the Application Enablement Services and the Communication Manager C-LAN is up.

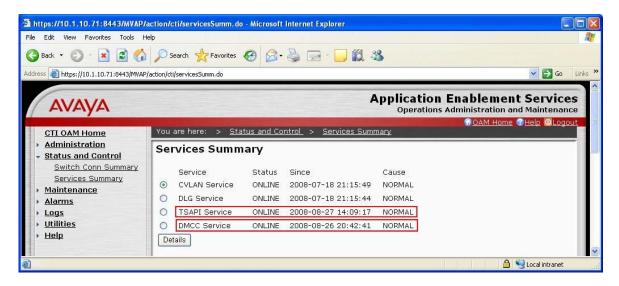
status aesvcs link				
	AE SERVICES	LINK STATUS		
Srvr/ AE Services Link Server	Remote IP	Remote Local Node Port	Msgs Sent	Msgs Rcvd
01/01 aes1	10. 1. 10. 71	32856 s8500-clan1	226	211

Verify the status of the administered TSAPI CTI link by using the **status aesvcs cti-link** command. The **Service State** field should display **established**.

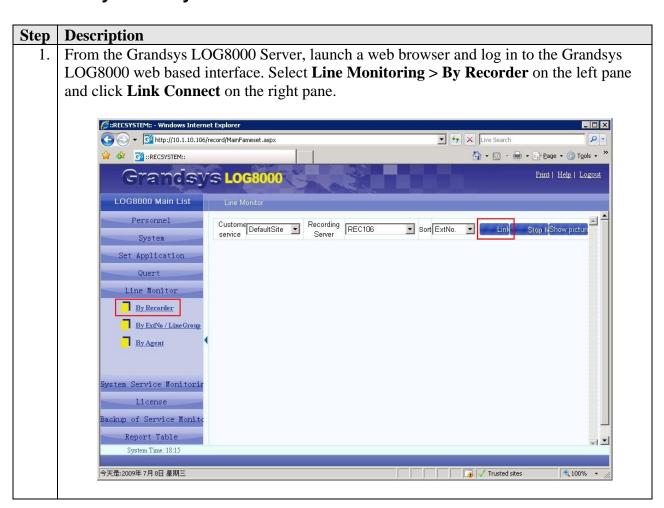
status	s aesvcs	cti-li	nk			
			AE SERVICES	CTI LINK STAT	US	
CTI Link	Version	Mnt Busy	AE Services Server	Service State	Msgs Sent	Msgs Rcvd
1	5	no	aes1	established	33	45

8.2. Verify Avaya Application Enablement Services

From the CTI OAM Admin web pages, verify the status of the TSAPI and DMCC Services by selecting **Status and Control > Services Summary** from the left pane. The **Status** field for both **TSAPI Service** and **DMCC Service** should display **ONLINE**.



8.3. Verify Grandsys LOG8000



Step | **Description**

2. Place a test call to an extension being recorded and verify that one of the recording stations on Grandsys LOG8000 becomes active as it records the call.



3. Query for the recording of the test call. Verify that the recording can be played back correctly.



9. Conclusion

These Application Notes illustrate the procedures for configuring Grandsys LOG8000 to monitor and record calls placed to and from stations and VDNs on Avaya AuraTM Communication Manager. In the configuration described in these Application Notes, Grandsys LOG8000 uses the DMCC Service of Avaya AuraTM Application Enablement Services to perform recording. All test cases were completed successfully.

10. Additional References

This section references the Avaya and Grandsys documentation that are relevant to these Application Notes.

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

- [1] Avaya MultiVantage® Application Enablement Services Administration and Maintenance Guide, Release 4.2, Document ID 02-300357, Issue 10, May 2008.
- [2] *Avaya Aura*[™] *Communication Manager Feature Description and Implementation*, Issue 7, May 2009, Document Number 555-245-205.

The following product documentation are available from Grandsys.

- [3] Grandsys LOG8000 System Installation Manual, Version 2.2.2, May 2009.
- [4] Grandsys LOG8000 System Operation Manual, Version 2.2.2, May 2009.

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