



Avaya Solution & Interoperability Test Lab

Application Notes for the Spok PC/PSAP, utilizing Spok CTI Layer, with Avaya Aura® Communication Manager and Avaya Aura® Application Enablement Services – Issue 1.0

Abstract

These Application Notes describe a compliance-tested configuration comprised of Avaya Aura® Communication Manager, Avaya Aura® Application Enablement Services, Avaya H.323 and Digital Telephones, and Spok PC/PSAP desktop applications.

Spok PC/PSAP is a Windows-based intelligent E911 workstation solution for a campus or municipality. Using the existing PBX telephone system as an “Automatic Number Identification (ANI)/Automatic Location Information (ALI) controller”, Spok PC/PSAP eliminates the need for external proprietary switching solutions and is able to perform all necessary telephony functions from the call taker’s PC keyboard. Spok PC/PSAP integrates with Spok CTI Layer, which is a middleware between Spok PC/PSAP and Avaya Aura® Application Enablement Services, to control and monitor phone states.

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 a compliance-tested configuration comprised of Avaya Aura® Communication Manager, Avaya Aura® Application Enablement Services, Avaya H.323 and Digital telephones, and Spok PC/PSAP applications.

Spok PC/PSAP is a PC and LAN based system used in a PSAP (Public Safety Answering Position – a physical location where 911 emergency telephone calls are received and then routed to the proper emergency services by the security agent or “911 operator” at the PSAP). Campuses or municipalities can set up a public or private PSAP using Spok PC/PSAP, which has the capabilities to extract ANI (Automatic Number Identification – phone number of the caller) from Emergency 911 trunks and retrieve corresponding ALI (Automatic Location Information – information about the call based on the ANI such as name, phone number, address, nearest cross street, etc.). Spok PC/PSAP integrates with Spok CTI Layer, which is a middleware between Spok PC/PSAP and Avaya Aura® Application Enablement Services, to control and monitor phone states.

It is the Spok CTI Layer service that actually uses the Avaya Aura® Application Enablement Services Device and Media Call Control (DMCC) Application Programming Interface (API) to share control of and monitor a physical telephone and receive the same terminal and first party call information received by the physical telephone. Spok PC/PSAP in turn uses the Spok CTI Layer service to control and monitor a physical telephone. Spok PC/PSAP registers as a DMCC station for Avaya H.323 or Avaya Digital endpoint configured on Communication Manager. The PC/PSAP applications regularly provide the database server with call and lamp state information concerning the controlled telephones.

2. General Test Approach and Test Results

The general approach was to exercise basic telephone and call operations on Avaya H.323 and Digital telephones using the aforementioned Spok desktop application.

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.

Avaya recommends our customers implement Avaya solutions using appropriate security and encryption capabilities enabled by our products. The testing referenced in these DevConnect Application Notes included the enablement of supported encryption capabilities in the Avaya products. Readers should consult the appropriate Avaya product documentation for further information regarding security and encryption capabilities supported by those Avaya products.

Support for these security and encryption capabilities in any non-Avaya solution component is the responsibility of each individual vendor. Readers should consult the appropriate vendor-supplied product documentation for more information regarding those products.

For the testing associated with these Application Notes, the interface between Avaya systems and Spok made use of secure DMCC.

2.1. Interoperability Compliance Testing

The interoperability compliance test included features and serviceability. The focus of the compliance test was primarily on verifying the interoperability between Spok PC/PSAP, Application Enablement Services, and Communication Manager. The main objectives were to verify that:

- The user may successfully use PC/PSAP to perform off-hook, on-hook, dial, answer, hold, retrieve, transfer, conference, and release operations on the physical telephone.
- Call from PC/PSAP to/from Avaya Endpoints (SIP, H.323 and Digital) and PSTN.
- The agent user may successfully use PC/PSAP to log into and out of an ACD, and move between agent work modes.
- Manual operations performed on the physical telephone are correctly reflected in the PC/PSAP GUI.
- PC/PSAP and manual telephone operations may be used interchangeably; for example, go off-hook using PC/PSAP and manually dial digits.
- Display and call information on the physical telephone is accurately reflected in the PC/PSAP GUI.
- Call states are consistent between PC/PSAP and the physical telephone.

Serviceability testing such as network failure and server reset for Spok PC/PSAP was also performed.

2.2. Test Results

All test cases were executed and passed with the exception of the observations below:

- During a scenario where the network connection from Spok PC/PSAP is lost, the CTI service on Spok PC/PSAP needed to be manually restarted to register the DMCC station again.
- Spok PC/PSAP does not support SIP endpoints configured on Communication Manager because it uses DMCC “Dependent” mode for device registrations, which is not supported for SIP telephones through the DMCC API.

2.3. Support

Technical support for the Spok PC/PSAP solution can be obtained by contacting Spok:

- URL – <http://www.spok.com>
- Phone – +1 (888) 797-7487

3. Reference Configuration

Figure 1 illustrates the configuration used in these Application Notes. The sample configuration shows an enterprise with an Application Enablement Services, Communication Manager, Media Server with an Avaya G450 Media Gateway. The PC/PSAP is configured to be in the same network as the enterprise. Endpoints include Avaya 9600 Series SIP, H.323 and Digital Telephones.

Note: Basic administration of Communication Manager and Application Enablement Services server is assumed. For details, see [1] and [2].

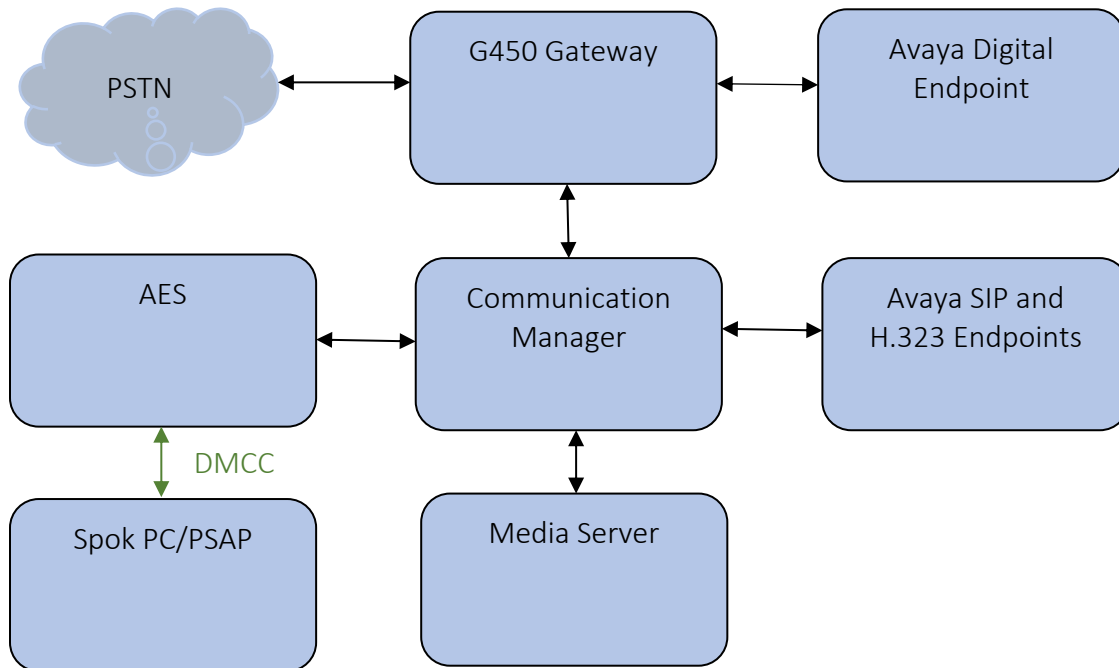


Figure 1: Spok PC/PSAP Test Configuration.

4. Equipment and Software Validated

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

Equipment	Software/Firmware
Avaya Aura® Communication Manager	8.0.1.1.0.822.25183
Avaya Aura® Application Enablement Services	8.0.1.0.2.5-0
Avaya Aura® Media Server	8.0.0.183
Avaya G450 Media Gateway	40.20.1
Avaya 9600 Series IP Telephones (H.323) <ul style="list-style-type: none">• 96x1 H.323• 96x1 SIP	<ul style="list-style-type: none">• 6.8102• 7.1.5.0
Avaya 9404 Digital Endpoint	2.0 SP8
Spok CTI Layer	7.x (7.0.0.6)
Spok PC/PSAP	11.x (11.11.0.404)

5. Configure Avaya Aura® Communication Manager

This section describes the procedures for configuring IP Services, Feature Access Codes, Abbreviated Dialing, and controlled telephones.

5.1. Configure IP Services

Enter the **change node-names ip** command. In the compliance-tested configuration, the **procr** IP address was used for registering H.323 endpoints, and for connectivity to Application Enablement Services.

```
change node-names ip
                                IP NODE NAMES
Name                            IP Address
aes15019                        10.64.150.19
aes8                             10.64.110.132
ams8                             10.64.110.136
cms18                            10.64.110.20
default                          0.0.0.0
egw1                             10.64.110.200
egw2                             10.64.110.201
procr                          10.64.110.131
procr6                           ::
sm8                              10.64.110.135
```

Enter the **change ip-services** command. On **Page 1**, configure the Service Type field to **AESVCS** and the Enabled field to **y**. The Local Node field should be pointed to the **procr** that was configured previously in the IP NODE NAMES form in this section. During the compliance test, the default port was used for the Local Port field.

```
change ip-services
                                Page 1 of 3
Service Enabled Local IP SERVICES Remote Remote
Type      y      Node Port      Node      Port
AESVCS   y    procr 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 **ssh**, and running the command **uname -a**. Enter an alphanumeric password for the Password field. Set the Enabled field to **y**. The same password will be configured on the Application Enablement Services server in **Section 6.2**.

```
change ip-services
                                Page 3 of 3
                                AE Services Administration
Server ID AE Services Password Enabled Status
          Server
1:      aes8          *          y          in use
2:
```

5.2. Configure Feature Access Codes (FAC)

Enter the **change feature-access-codes** command. On **Page 1** of the FEATURE ACCESS CODE (FAC) form, verify the **Auto Route Selection (ARS) – Access Code 1** field is set to **9**.

```
change feature-access-codes                                     Page 1 of 11
                                                              FEATURE ACCESS CODE (FAC)
Abbreviated Dialing List1 Access Code:
Abbreviated Dialing List2 Access Code:
Abbreviated Dialing List3 Access Code:
Abbreviated Dial - Prgm Group List Access Code:
Announcement Access Code:
Answer Back Access Code: #25
Attendant Access Code:
Auto Alternate Routing (AAR) Access Code: 8
Auto Route Selection (ARS) - Access Code 1: 9      Access Code 2:
Automatic Callback Activation:                      Deactivation:
Call Forwarding Activation Busy/DA: *97 All: *99    Deactivation: *98
```

5.3. Configure Dialplan

Enter the **change dialplan analysis** command. Create a single digit dial string with **9** and associate it with **Feature Access Code (fac)**.

```
change dialplan analysis                                     Page 1 of 12
                                                              DIAL PLAN ANALYSIS TABLE
                                                              Location: all                               Percent Full: 1
Dialed   Total Call   Dialed   Total Call   Dialed   Total Call
String   Length Type   String   Length Type   String   Length Type
1        3    dac        1        3    dac        1        3    dac
1        4    ext        1        4    ext        1        4    ext
1        5    ext        1        5    ext        1        5    ext
3        10   ext        3        10   ext        3        10   ext
8        1    fac        8        1    fac        8        1    fac
9        1    fac        9        1    fac        9        1    fac
*        3    dac        *        3    dac        *        3    dac
#        3    dac        #        3    dac        #        3    dac
```

5.4. Configure Hunt Group

Enter the **add hunt-group n** command, where **n** is an unused hunt group number. On **Page 1** of the HUNT GROUP form, assign a descriptive Group Name and Group Extension valid in the provisioned dial plan. Set **ACD**, **Queue** and **Vector** to **y**.

```
add hunt-group 1                                     Page 1 of 4
                                     HUNT GROUP
Group Number: 1                                     ACD? y
Group Name: Skill 1                                 Queue? y
Group Extension: 59001                             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. Configure Abbreviated Dialing

Enter the **add abbreviated-dialing system** command. In the **DIAL CODE** list, enter the Feature Access Codes for ACD Login and Logout.

```
change abbreviated-dialing system                   Page 1 of 1
                                     ABBREVIATED DIALING LIST
                                     SYSTEM LIST
Size (multiple of 5): 5      Privileged? n      Label Language:english
DIAL CODE                    LABELS (FOR STATIONS THAT DOWNLOAD LABELS)
01: *01                      01: Log-in
02: *06                      02: Log-out
03:                          03: *****
04:                          04: *****
05:                          05: *****
```


5.6. Configure Controlled Telephones

Enter the **change station r** command, where **r** is the extension of a registered, physical Avaya H.323 or Digital telephone. On **Page 1** of the **station** form, enter a phone Type, descriptive name, Security Code and set **IP SoftPhone** field to **y** to allow the physical station to be controlled by a softphone such as the Spok PC/PSAP application.

```
change station 57001                                     Page 1 of 5
                                                    STATION
Extension: 57001                                     Lock Messages? n          BCC: 0
  Type: 9641                                           Security Code: *          TN: 1
  Port: S00039                                         Coverage Path 1:          COR: 1
  Name: Spok PCPSAP 1                               Coverage Path 2:          COS: 1
Unicode Name? n                                       Hunt-to Station:          Tests? y
STATION OPTIONS
                                                    Time of Day Lock Table:
  Loss Group: 19                                       Personalized Ringing Pattern: 1
                                                    Message Lamp Ext: 57001
  Speakerphone: 2-way                                   Mute Button Enabled? y
  Display Language: english                             Button Modules: 0
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** of the station form, for **ABBREVIATED DIALING List 1**, enter the abbreviated dialing group configured in previous section. On **Pages 4, 5, and 6** of the station forms, configure the following **BUTTON ASSIGNMENTS** in addition to the call-appr (call appearance) buttons as shown below:

```

change station 57001                                     Page 4 of 5
                                     STATION
SITE DATA
  Room:                               Headset? n
  Jack:                               Speaker? n
  Cable:                              Mounting: d
  Floor:                              Cord Length: 0
  Building:                            Set Color:

ABBREVIATED DIALING
  List1: system          List2:          List3:

BUTTON ASSIGNMENTS
  1: call-appr          5: call-pkup
  2: call-appr          6: next
  3: brdg-appr B:1 E:57002 7: aux-work RC: Grp:
  4: brdg-appr B:1 E:57003 8: auto-in   Grp:

change station 11054                                     Page 5 of 7
                                     STATION
BUTTON ASSIGNMENTS
  9: abrv-dial List: 1 DC: 01 HL? n
 10: abrv-dial List: 1 DC: 02 HL? n
 11: release
 12: togle-swap

change station 57001                                     Page 5 of 5
                                     STATION
BUTTON ASSIGNMENTS
  9: abrv-dial List: 1 DC: 01
 10: abrv-dial List: 1 DC: 02
 11: release
 12: togle-swap
 13:
 14:
 15:
 16:
 17: q-calls Grp: 2

```

Repeat the instructions provided in this section for each physical station that is to be controlled / monitored by the Spok CTI Layer.

6. Configure Application Enablement Services

The Application Enablement Services server enables Computer Telephony Interface (CTI) applications to control and monitor telephony resources on Communication Manager.

This section assumes that installation and basic administration of the Application Enablement Services server has been performed. The steps in this section describe the configuration of a Switch Connection, a CTI user, a DMCC port.

6.1. Device and Media Call Control API Station Licenses

The Spok PC/PSAP Service instances appear as “virtual” stations/softphones to Communication Manager. Each of these virtual stations, hereafter called Device and Media Call Control API station, requires a license. Note that this is separate and independent of Avaya IP Softphone licenses, which are required for Avaya IP Softphones but not required for Device and Media Call Control API stations. To check and verify that there are sufficient DMCC licenses, log in to <https://<IP address of the Application Enablement Services server>/index.jsp>, and enter appropriate login credentials to access the Application Enablement Services Management Console page.

Select the **Licensing** → **WebLM Server Access** link from the left pane of the window (not shown). During the compliance testing, Avaya Aura System Manager was used as a license server.

Provide appropriate login credentials and log in.

Recommended access to System Manager is via FQDN.
[Go to central login for Single Sign-On](#)

If IP address access is your only option, then note that authentication will fail in the following cases:

- First time login with "admin" account
- Expired/Reset passwords

Use the "Change Password" hyperlink on this page to change the password manually, and then login.

Also note that single sign-on between servers in the same security domain is not supported when accessing via IP address.

User ID:

Password:

[Change Password](#)

Supported Browsers: Internet Explorer 11.x or Firefox 59.0, 60.0 and 61.0.

Navigate to **Home → Services → Licenses**. On the WebLM Home page, select **License Products → Application_Enablement** link from the left pane of the window.

On the Licensed Features page, verify that there are sufficient DMCC licenses.

Note: TSAPI licenses (1 per agent station) are also required if calls routed to agent stations via ACD. Without TSAPI licenses, the agents will not see the First Party Call Control (1PCC) calling party information. i.e., Calling Party Number.

WebLM Home	Application Enablement (CTI) - Release: 8 - SID: 10503000 Standard
Install license	You are here: Licensed Products > Application_Enablement > View License Capacity
Licensed products	License installed on: October 31, 2018 11:09:07 AM -06:00
APPL_ENAB	
▼ Application_Enablement	
View license capacity	<input type="text" value="License File Host IDs"/>
View peak usage	
ASBCE	<input type="text" value="Licensed Features"/>
▶ Session_Border_Controller_E_AE	
CE	13 Items <input type="button" value="Show"/> All <input type="button" value="▼"/>
▶ COLLABORATION_ENVIRONMENT	
CMM	
▶ Communication_Manager_Messaging	
Configure Centralized Licensing	
COMMUNICATION_MANAGER	
▶ Call_Center	

Feature (License Keyword)	Expiration date	Licensed capacity
Device Media and Call Control VALUE_AES_DMCC_DMC	permanent	1000
AES ADVANCED LARGE SWITCH VALUE_AES_AEC_LARGE_ADVANCED	permanent	8
AES HA LARGE VALUE_AES_HA_LARGE	permanent	8
AES ADVANCED MEDIUM SWITCH VALUE_AES_AEC_MEDIUM_ADVANCED	permanent	8

6.2. Configure Switch Connection

Launch a web browser, enter `https://<IP address of the Application Enablement Services server>` in the address field, and log in with the appropriate credentials for accessing the Application Enablement Services Management Console pages.



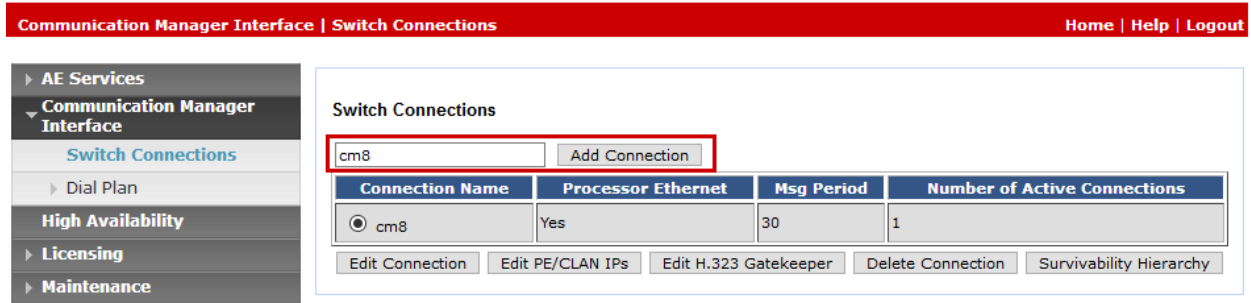
Application Enablement Services Management Console

Please login here:

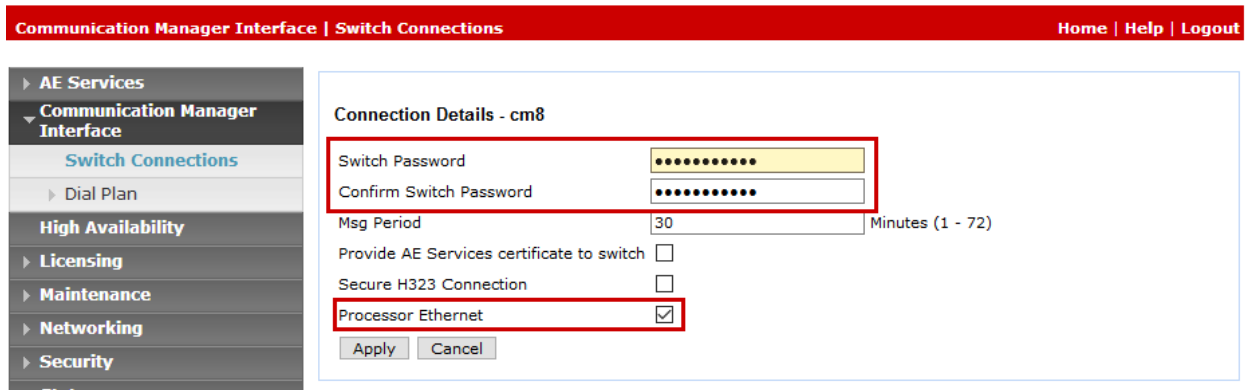
Username

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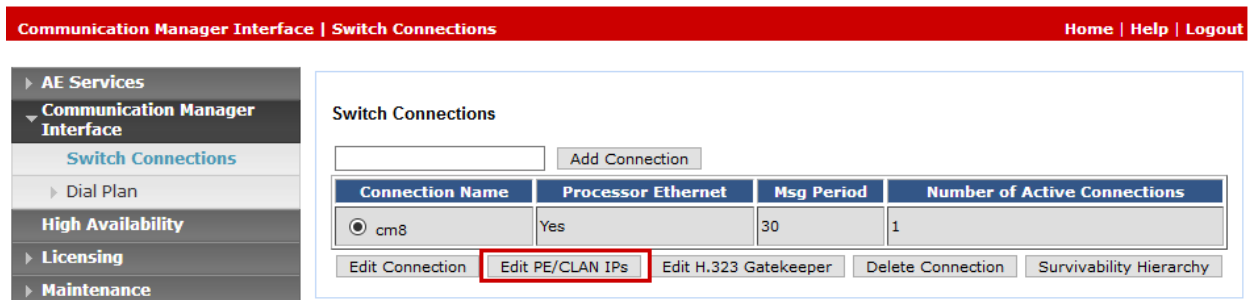
Click on **Communication Manager Interface** → **Switch Connection** in the left pane to invoke the Switch Connections page. A Switch Connection defines a connection between the Application Enablement Services and Communication Manager. Enter a descriptive name for the switch connection and click on **Add Connection**.



The next window that appears prompts for the **Switch Password**. Enter the same password that was administered in Communication Manager in **Section 5.1**. Check box for **Processor Ethernet**. Click on **Apply**.



After returning to the Switch Connections page, select the radio button corresponding to the switch connection added previously, and click on **Edit PE/CLAN IPs**.



Enter the IP address of Procr used for Application Enablement Services connectivity from **Section 5.1**, and click on **Add Name or IP**.

Communication Manager Interface | Switch Connections Home | Help | Logout

AE Services
 Communication Manager Interface
 Switch Connections
 Dial Plan
 High Availability
 Licensing
 Maintenance

Edit Processor Ethernet IP - cm8

10.64.110.131

Name or IP Address	Status
10.64.110.131	In Use

After returning to the Switch Connections page, select the radio button corresponding to the switch connection added previously, and click on the **Edit H.323 Gatekeeper** button.

Communication Manager Interface | Switch Connections Home | Help | Logout

AE Services
 Communication Manager Interface
 Switch Connections
 Dial Plan
 High Availability
 Licensing
 Maintenance

Switch Connections

Connection Name	Processor Ethernet	Msg Period	Number of Active Connections
<input checked="" type="radio"/> cm8	Yes	30	1

On the **Edit H.323 Gatekeeper – acm** page, enter the procr IP address which will be used for the DMCC service. Click on **Add Name or IP**.

Communication Manager Interface | Switch Connections Home | Help | Logout

AE Services
 Communication Manager Interface
 Switch Connections
 Dial Plan
 High Availability
 Licensing

Edit H.323 Gatekeeper - cm8

10.64.110.131

Name or IP Address

10.64.110.131

6.3. Configure the CTI Users

Navigate to **User Management** → **User Admin** → **Add User** link from the left pane of the window. On the Add User page, provide the following information:

- User Id
- Common Name
- Surname
- User Password
- Confirm Password

Select **Yes** using the drop-down menu on the CT User field. This enables the user as a CTI user. Default values may be used in the remaining fields. Click the **Apply** button (not shown) at the bottom of the screen to complete the process.

The screenshot shows the 'Add User' form in a web application. The breadcrumb navigation at the top reads 'User Management | User Admin | Add User'. The left sidebar shows a tree view with 'User Management' expanded to 'User Admin', where 'Add User' is selected. The main form area contains the following fields:

- * User Id: spok
- * Common Name: spok
- * Surname: spok
- * User Password: [masked]
- * Confirm Password: [masked]
- Admin Note: [empty]
- Avaya Role: None (dropdown)
- Business Category: [empty]
- Car License: [empty]
- CM Home: [empty]
- Css Home: [empty]
- CT User: Yes (dropdown)
- Department Number: [empty]

The above information (User ID and User Password) must match with the information configured in the Spok PC/PSAP Configuration page in **Section 7**.

Once the user is created, navigate to the **Security** → **Security Database** → **CTI Users** → **List All Users** link from the left pane of the window. Select the User ID created previously, and click the **Edit** button to set the permission of the user (not shown).

Provide the user with unrestricted access privileges by checking the **Unrestricted Access** checkbox. Click on the **Apply Changes** button.

Edit CTI User		
User Profile:	User ID	spok
	Common Name	spok
	Worktop Name	NONE
	Unrestricted Access	<input checked="" type="checkbox"/>
Call and Device Control:	Call Origination/Termination and Device Status	None
Call and Device Monitoring:	Device Monitoring	None
	Calls On A Device Monitoring	None
	Call Monitoring	<input type="checkbox"/>
Routing Control:	Allow Routing on Listed Devices	None
<input type="button" value="Apply Changes"/> <input type="button" value="Cancel Changes"/>		

6.4. Configure the DMCC Port

Navigate to the **Networking** → **Ports** link, from the left pane of the window, to set the DMCC server port. During the compliance test, the default port values were utilized. The following screen displays the default port values. Since the unencrypted port was utilized during the compliance test, set the Unencrypted Port field to **Enabled**. Default values may be used in the remaining fields. Click the **Apply Changes** button (not shown) at the bottom of the screen to complete the process.

Ports		
CVLAN Ports	Unencrypted TCP Port	9999 <input type="radio"/> Enabled <input type="radio"/> Disabled
	Encrypted TCP Port	9998 <input type="radio"/> Enabled <input type="radio"/> Disabled
DLG Port	TCP Port	5678
TSAPI Ports	TSAPI Service Port	450 <input type="radio"/> Enabled <input type="radio"/> Disabled
	Local TLINK Ports	
	TCP Port Min	1024
	TCP Port Max	1039
	Unencrypted TLINK Ports	
	TCP Port Min	1050
	TCP Port Max	1065
Encrypted TLINK Ports		
TCP Port Min	1066	
TCP Port Max	1081	
DMCC Server Ports	Unencrypted Port	4721 <input checked="" type="radio"/> Enabled <input type="radio"/> Disabled
	Encrypted Port	4722 <input type="radio"/> Enabled <input type="radio"/> Disabled
	TR/87 Port	4723 <input type="radio"/> Enabled <input checked="" type="radio"/> Disabled

7. Configure Spok PC/PSAP

Spok installs, configures, and customizes the PC/PSAP applications for their end customers. Spok PC/PSAP integrates with Spok CTI Layer, which is a middleware between Spok PC/PSAP and Application Enablement Services, to control and monitor the phone states. Thus, only the Spok CTI layer will be discussed in these Application Notes.

Note: Avaya phones as the network supplier for the agent workstations is not supported by Spok. Agent workstations should have their own network connection, separate from Avaya phones.

The following shows the **Spok AES CTI Services Setup** page. Provide the following information:

Under DMCC Settings

- **AES Server** – Enter the IP address of the Application Enablement Service.
- **Switch IP Address** – Enter the procr IP address of Communication Manager.
- **Port** – Enter the DMCC port (4721).
- **User** – Enter the user name created for Spok PC/PSAP in **Section 6.3**.
- **Password** – Enter the password created for Spok PC/PSAP in **Section 6.3**.

Under Phone Device Settings

- **Extension** – Enter the extension that will be controlled by the Spok PC/PSAP.
- **Security Code** – Enter the security code for the controlled station.
- **Release Button** – Enter the Release button assigned for the controlled station.
- **Line Appearances** – Enter the line appearances used for the controlled station.

DMCC Settings:

AES Server:	10.64.110.132
Switch Name:	cm8
Switch IP Interface:	10.64.110.131
Port:	Secure (4722)
Application Id:	13
Device Instance:	0
Local Certificate File:	C:\Users\spokpsapa\Desktop\SystemManagerCA.crt
SSL Protocol:	TLSv1 (Transport Layer Security version 1)
User (default = cmapi):	spok
Password:	*****
Media Mode:	No Media
Shared Control:	False
Dependency Mode:	Dependent
AES Version:	7.0
Telecomuter Extension:	
<input type="checkbox"/>	Monitor Call Information
<input type="checkbox"/>	Monitor Media Device
<input type="checkbox"/>	Monitor Device Service

Phone Device Settings:

Extension:	57001	RLT Transfer Button Id:	
Security Code:	*****	Release Button Id:	11
Max SCA Timer (ms):	250	Toggle-Swap Button Id:	
<input type="checkbox"/>	Press Release Button Upon Cancel		
Park Access Code:			
Unpark Access Code:			

Line Appearances:

Line 1	Button Id = 1	Display Id = a
Line 2	Button Id = 2	Display Id = b
Line 3	Button Id = 3	Display Id = c BRIDGE
Line 4	Button Id = 4	Display Id = d BRIDGE

Service Settings:

Listener Port:	973
Home Directory:	D:\spok\Applications\PcPsap\CTI_Service
Configuration File Name:	D:\spok\Applications\PcPsap\CTI_Service\CTI_Service\c...
DLL File Name:	D:\spok\Applications\PcPsap\CTI_Service\bin\amcom_cr...
LUA Agent Function File:	D:\spok\Applications\PcPsap\CTI_Service\CTI_Service\c...
LUA Agent State File:	D:\spok\Applications\PcPsap\CTI_Service\CTI_Service\c...
LUA App Specific File:	D:\spok\Applications\PcPsap\CTI_Service\CTI_Service\c...
<input checked="" type="checkbox"/>	Send SCA = 0 at the beginning of call state messages

Debug Settings:

File Name:	trace				
Number of Files:	10	File Size:	10000		
Directory:	D:\spok\Applications\PcPsap\CTI_Service\trace				
<input checked="" type="checkbox"/>	Level 1	<input checked="" type="checkbox"/>	Level 16	<input checked="" type="checkbox"/>	Level 256
<input checked="" type="checkbox"/>	Level 2	<input checked="" type="checkbox"/>	Level 32	<input checked="" type="checkbox"/>	Level 512
<input checked="" type="checkbox"/>	Level 4	<input checked="" type="checkbox"/>	Level 64	<input checked="" type="checkbox"/>	Level 1024
<input checked="" type="checkbox"/>	Level 8	<input checked="" type="checkbox"/>	Level 128	<input checked="" type="checkbox"/>	Level 2048

8. Verification Steps

The following steps may be used to verify the configuration:

- From the Spok client computers, ping IP interfaces, in particular the Application Enablement Services server, and verify connectivity.
- For the physical IP telephones, verify that the physical telephones are registered by using the **list registered-ip-stations** command on the Communication Manager System Access Terminal (SAT). For the physical Digital telephones, verify that the telephones are attached to the correct ports.
- Go off-hook and on-hook on the controlled telephones manually and use PC/PSAP to verify consistency.
- Place and answer calls from the controlled telephones manually and use PC/PSAP to verify consistency.

9. Conclusion

These Application Notes described a compliance-tested configuration comprised of Communication Manager, Application Enablement Services, Avaya H.323 and Digital telephones, and the Spok PC/PSAP application. Spok PC/PSAP allows a user to operate a physical telephone and view call and telephone display information through a graphical user interface (GUI). During compliance testing, calls were successfully placed to and from Avaya H.323 and Digital telephones that were controlled and monitored by the Spok PC/PSAP application.

10. Additional References

Product documentation for Avaya products may be found at <http://support.avaya.com>.

[1] *Administering Avaya Aura® Communication Manager, Release 8.0.1*

[2] *Administering Avaya Aura® Avaya Aura® Application Enablement Services, Release 8.0.1*,

Product information for Spok products may be found at <http://www.spok.com>.

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