



## Avaya Solution & Interoperability Test Lab

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# Application Notes for Configuring Presence Technology Presence Suite R10.1 with Avaya Aura® Communication Manager R7.0 and Avaya Aura® Application Enablement Services R7.0 – Issue 1.0

### Abstract

These Application Notes describe the configuration steps for Presence Technology Presence Suite to successfully interoperate with Avaya Aura® Communication Manager and Avaya Aura® Application Enablement Services. Presence Suite is a multi-channel contact management suite which handles voice, text chat, email and web contact mechanisms. Presence Suite integrates with the Avaya solution by using the Telephony Services Application Programmer Interface (TSAPI) provided by Avaya Aura® Application Enablement Services to monitor and control agent stations, and handle routing of external calls.

Readers should pay attention to **section 2**, in particular the scope of testing as outlined in **Section 2.1** as well as the 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 compliance tested configuration using Presence Suite R10.1 and Avaya Aura® Communication Manager R7.0 with Avaya Aura® Application Enablement Services R7.0 (AES). Presence Suite is a multi-channel contact management suite able to handle voice, e-mail and web chat contact mechanisms. The Telephony Services Application Programmer Interface (TSAPI) provided by Avaya Aura® Application Enablement Services is used to monitor and control agent stations, generate phantom calls for non-voice contacts and handle routing of external calls. Presence Suite consists of a number of modules. Only the following modules were tested:

- Presence Voice Outbound
- Presence Voice Inbound
- Presence Mail Interactions
- Presence Web Interactions

Link Failure\Recovery was also tested to ensure successful reconnection on link failure. Upon starting the Presence Server application, the application automatically queries Avaya Aura® Application Enablement Services for device status and requests monitoring. The Presence Server specifies where to route each call and hence how to handle the calls, based on agent status information that the application tracks from CTI device query results and event reports received from Avaya Aura® Application Enablement Services.

## 2. General Test Approach and Test Results

Testing included validating the correct operation of typical contact centre functions including, inbound and outbound service calls. Functionality testing included basic telephony operations such as answer, hold/retrieve, transfer, and conference. This was carried out for the inbound and outbound service calls. Email, Web call back and Web chat were also tested. Additional features such as call capturing, direct agent transfer and malicious calls were tested. The serviceability test cases were performed manually by busying out and releasing the CTI link and by disconnecting and reconnecting LAN cables.

**Note:** Only call control of H.323 stations is possible, no SIP stations were used in the compliance testing.

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 both feature functionality and serviceability testing. The feature functionality testing focused on verifying Presence Suite handling of TSAPI messages in the areas of routing, call control and event notification. The serviceability testing focused on verifying the Presence Suite ability to recover from adverse conditions, such as stopping the TSAPI Service, taking the CTI link offline and disconnecting the Ethernet cable from all the devices in the solution.

## 2.2 Test Results

All test cases passed successfully.

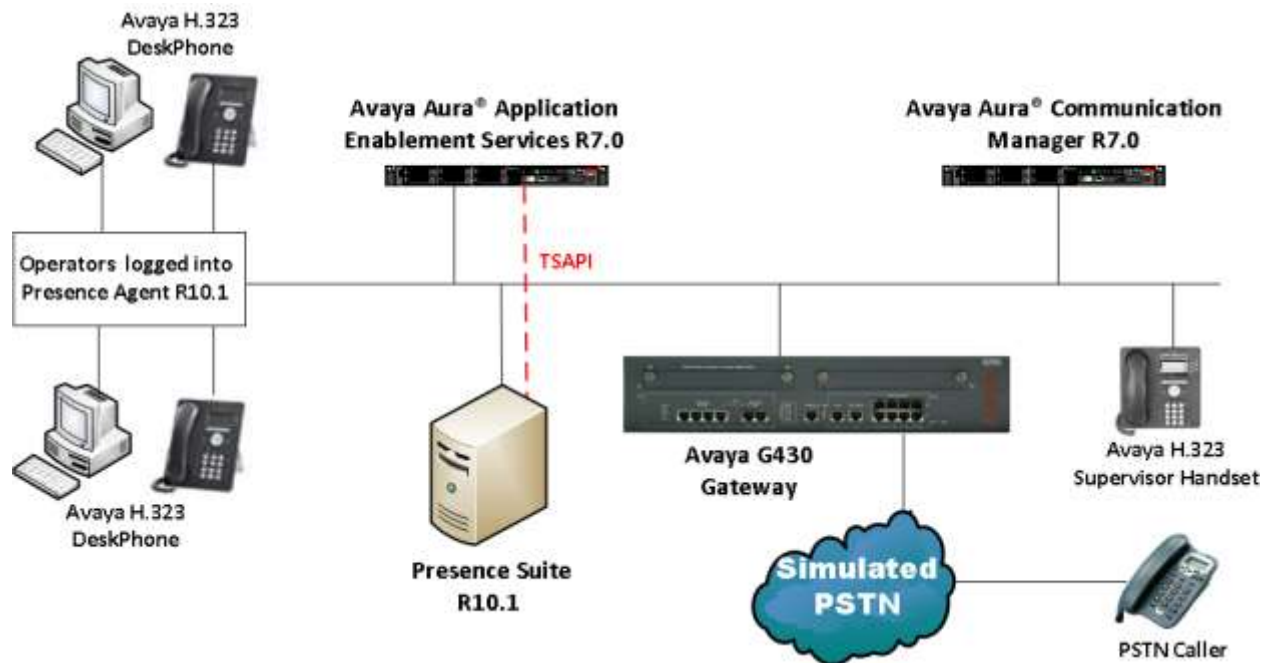
## 2.3 Support

Technical support can be obtained for Presence Technology Presence Suite as follows:

- Email: [support@presenceco.com](mailto:support@presenceco.com)
- Website: [www.presenceco.com](http://www.presenceco.com)
- Phone: +34 93 10 10 300

### 3. Reference Configuration

**Figure 1** shows the network topology during interoperability testing. A Communication Manager with an Avaya G430 Media Gateway was used as the hosting PBX. Presence Suite, including Presence Agent PC's, are connected to the LAN and controls the Avaya H323 IP telephones via Application Enablement Services using TSAPI.



**Figure 1: Avaya Aura® Communication Manager R7.0 with Aura® Application Enablement Services R7.0 and Presence Technology Presence Suite Server configuration**

## 4. Equipment and Software Validated

All the hardware and associated software used in the compliance testing is listed below.

Equipment/Software	Release/Version
Avaya Aura® Communication Manager running on Avaya S8800 Server	R7.0 SP1 Revision 7.0.0.1.0.441.22438
Avaya Aura® Application Enablement Services running on Avaya S8800 Server	R7.0 Build No – 7.0.0.0.0.13-0
Avaya G430 Gateway	R37.19.0
Avaya 96xx Series Deskphone	96xx H.323 Release 3.2 SP3
Presence Server running on Windows Server 2008 R2	R10.1
Presence Client running on Windows 7 SP1 and Windows Server 2008 R2	R10.1

## 5. Configure Avaya Aura® Communication Manager

The information provided in this section describes the configuration of Communication Manager for this solution. For all other provisioning information such as initial installation and configuration, please refer to the product documentation in **Section 10**. The configuration and verification operations illustrated in this section were all performed using Communication Manager System Administration Terminal (SAT). The configuration operations described in this section can be summarized as follows:

- Verify System Features
- Administer SIT Treatment for Call Classification
- Define Feature Access Codes (FAC)
- Administer Trunk Group
- Administer Hunt Groups, Vectors and VDN's
- Administer Class of Restriction
- Administer Agent Logins
- Administer Agent Stations
- Administer Phantom Stations
- Note procr IP Address for AES Connectivity
- Configure Transport link for AES Connectivity
- Configure CTI Link for TSAPI Service

### 5.1 Verify System Features

Use the **display system-parameters customer-options** command to verify that Communication Manager has permissions for features illustrated in these Application Notes. On **Page 3**, ensure that **Computer Telephony Adjunct Links?** is set to **y** and **Answer Supervision by Call Classifier?** is set to **y** as shown below.

display system-parameters customer-options		Page 3 of 11
OPTIONAL FEATURES		
Abbreviated Dialing Enhanced List? y	Audible Message Waiting? y	
Access Security Gateway (ASG)? n	Authorization Codes? y	
Analog Trunk Incoming Call ID? y	CAS Branch? n	
A/D Grp/Sys List Dialing Start at 01? y	CAS Main? n	
<b>Answer Supervision by Call Classifier? y</b>	Change COR by FAC? n	
ARS? y	<b>Computer Telephony Adjunct Links? y</b>	
ARS/AAR Partitioning? y	Cvg Of Calls Redirected Off-net? y	
ARS/AAR Dialing without FAC? y	DCS (Basic)? y	
ASAI Link Core Capabilities? n	DCS Call Coverage? y	
ASAI Link Plus Capabilities? n	DCS with Rerouting? y	
Async. Transfer Mode (ATM) PNC? n	Digital Loss Plan Modification? y	
Async. Transfer Mode (ATM) Trunking? n	DS1 MSP? y	
ATM WAN Spare Processor? n	DS1 Echo Cancellation? y	
ATMS? y		
Attendant Vectoring? y		

On **Page 6**, verify the following customer options are set to **y** as shown below.

- **ACD?** to **y**
- **Vectoring (Basic)?** to **y**
- **Expert Agent Selection (EAS)?** to **y**

<b>display system-parameters customer-options</b>		<b>Page 6 of 11</b>
CALL CENTER OPTIONAL FEATURES		
Call Center Release: 7.0		
<b>ACD? y</b>	Reason Codes? y	
BCMS (Basic)? y	Service Level Maximizer? n	
BCMS/VuStats Service Level? y	Service Observing (Basic)? y	
BSR Local Treatment for IP & ISDN? y	Service Observing (Remote/By FAC)? y	
Business Advocate? n	Service Observing (VDNs)? y	
Call Work Codes? y	Timed ACW? y	
DTMF Feedback Signals For VRU? y	<b>Vectoring (Basic)? y</b>	
Dynamic Advocate? n	Vectoring (Prompting)? y	
<b>Expert Agent Selection (EAS)? y</b>	Vectoring (G3V4 Enhanced)? y	
EAS-PHD? y	Vectoring (3.0 Enhanced)? y	
Forced ACD Calls? n	Vectoring (ANI/II-Digits Routing)? y	
Least Occupied Agent? y	Vectoring (G3V4 Advanced Routing)? y	
Lookahead Interflow (LAI)? y	Vectoring (CINFO)? y	
Multiple Call Handling (On Request)? y	Vectoring (Best Service Routing)? y	
Multiple Call Handling (Forced)? y	Vectoring (Holidays)? y	
PASTE (Display PBX Data on Phone)? y	Vectoring (Variables)? y	

Use the command **display system-parameters features** and on **Page 1**, verify that the **Trunk-to-Trunk Transfer** option is set to **all** as shown below.

<b>display system-parameters features</b>	<b>Page 1 of 20</b>
FEATURE-RELATED SYSTEM PARAMETERS	
Self Station Display Enabled? n	
<b>Trunk-to-Trunk Transfer: all</b>	
Automatic Callback with Called Party Queuing? n	
Automatic Callback - No Answer Timeout Interval (rings): 3	
Call Park Timeout Interval (minutes): 10	
Off-Premises Tone Detect Timeout Interval (seconds): 20	
AAR/ARS Dial Tone Required? y	
Music (or Silence) on Transferred Trunk Calls? no	
DID/Tie/ISDN/SIP Intercept Treatment: attendant	
Internal Auto-Answer of Attd-Extended/Transferred Calls: transferred	
Automatic Circuit Assurance (ACA) Enabled? n	

On **page 10** ensure that **Station Tone Forward Disconnect** is set to **silence** as shown below.

```
display system-parameters features                                     Page 10 of 20
                                FEATURE-RELATED SYSTEM PARAMETERS

                                Pull Transfer: n                      Update Transferred Ring Pattern? n
                                Outpulse Without Tone? y              Wait Answer Supervision Timer? n
                                Misoperation Alerting? n              Repetitive Call Waiting Tone? n
                                Allow Conference via Flash? y
                                Vector Disconnect Timer (min):        Network Feedback During Tone Detection? y
                                Hear Zip Tone Following VOA? y        System Updates Time On Station Displays? n

                                Station Tone Forward Disconnect: silence
                                Level Of Tone Detection: precise
                                Charge Display Update Frequency (seconds): 30

                                Onhook Dialing on Terminals? n
                                Edit Dialing on 96xx H.323 Terminals? n
                                Allow Crisis Alert Across Tenants? n
                                Send DTMF Over Telecommuter Link? y

ITALIAN DCS PROTOCOL
Italian Protocol Enabled? n
```

Use the command **display system-parameters features** and on **Page 11**, verify that the **Expert Agent Selection (EAS) Enabled?** option is set to **y** as shown below.

```
display 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
```



On page 12 ensure that **ACW Agents Considered Idle** is set to **y**.

```
display system-parameters features                                     Page 12 of 20
                                FEATURE-RELATED SYSTEM PARAMETERS

AGENT AND CALL SELECTION
    MIA Across Splits or Skills? n
    ACW Agents Considered Idle? y
    Call Selection Measurement: current-wait-time
    Service Level Supervisor Call Selection Override? n
    Auto Reserve Agents: none
    Block Hang-up by Logged-in Auto-Answer Agents? n

CALL MANAGEMENT SYSTEM
    REPORTING ADJUNCT RELEASE (determines protocol used by appl link)
        CMS (appl mis):
        AAPC/IQ (appl ccr):

        BCMS/VuStats LoginIDs? y
        BCMS/VuStats Measurement Interval: hour
        BCMS/VuStats Abandon Call Timer (seconds):
        Validate BCMS/VuStats Login IDs? n
        Clear VuStats Shift Data: on-login
        Remove Inactive BCMS/VuStats Agents? n
```

On Page 13, verify that **Call Classification After Answer Supervision** option is set to **y** as shown below.

```
display 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

    Interruptible Aux Notification Timer (sec): 3

ASAI
    Copy ASAI UUI During Conference/Transfer? y
    Call Classification After Answer Supervision? y
    Send UCID to ASAI? y
    For ASAI Send DTMF Tone to Call Originator? y
```

## 5.2 Administer Special Information Tones Treatment for Call Classification

This form is used to specify the treatment of Special Information Tones (SIT) used for outbound call management type calls with USA tone characteristics. Enter the **change sit-treatment** command. Set the **Pause Duration** to **0.8** and **Talk Duration** to **3.0**. Please note this may vary depending on the country where the PBX is installed.

<b>change sit-treatment</b>	Page 1 of 1
SIT TREATMENT FOR CALL CLASSIFICATION	
SIT Ineffective Other: dropped	
SIT Intercept: dropped	
SIT No Circuit: dropped	
SIT Reorder: dropped	
SIT Vacant Code: dropped	
SIT Unknown: dropped	
AMD Treatment: dropped	
<b>Pause Duration (seconds): 0.8</b>	
<b>Talk Duration (seconds): 3.0</b>	

## 5.3 Administer Trunk

The PSTN trunk may differ depending on the installation for the compliance testing a QSIG ISDN trunk was used and configured in the following way. Use the **change trunk-group n** command, where **n** is the trunk group number for the pre-configured ISDN trunk which will be used for inbound and outbound service calls. It is assumed that the ISDN trunk and the corresponding signaling group are already configured. The trunk group number used for interoperability testing is **9**. On **Page 1** set the **COR** (class of restriction) to **1**, this is the COR used for the sample configuration.

<b>change trunk-group 9</b>	Page 1 of 22	
TRUNK GROUP		
Group Number: 1	Group Type: isdn	CDR Reports: y
Group Name: Simulated PSTN	<b>COR: 1</b>	TN: 1 TAC: *19
Direction: two-way	Outgoing Display? y	Carrier Medium: PRI/BRI
Dial Access? y	Busy Threshold: 255	Night Service:
Queue Length: 0		
Service Type: tie	Auth Code? n	TestCall ITC: rest
	Far End Test Line No:	
TestCall BCC: 4		

On **Page 3**, **UUI IE Treatment** was set to **service-provider**. Default values may be used in the remaining fields.

change trunk-group 2		Page 3 of 22
TRUNK FEATURES		
ACA Assignment? n	Measured: none	Wideband Support? n
		Maintenance Tests? y
	Data Restriction? n	NCA-TSC Trunk Member:
	Send Name: n	Send Calling Number: n
Used for DCS? n		Send EMU Visitor CPN? n
Suppress # Outpulsing? n	Format: public	
Outgoing Channel ID Encoding: preferred	<b>UUI IE Treatment: service-provider</b>	

## 5.4 Administer Hunt Groups, Call Vectors and Vector Directory Numbers

In order for calls to be routed to agents, Hunt Groups (skills), Vectors, and Vector Directory Numbers (VDN) must be configured.

### 5.4.1 Hunt Groups

Enter the **add hunt-group n** command where **n** in the example below is **33**. On **Page 1** of the **hunt-group** form, assign a **Group Name** and **Group Extension** valid under the provisioned dial plan. Set the following options to **y** as shown below.

- **ACD?** to **y**
- **Queue?** to **y**
- **Vector?** to **y**

add hunt-group 33		Page 1 of 4
HUNT GROUP		
Group Number: 33	ACD? y	
Group Name: Presenceco Inbound	Queue? y	
Group Extension: 8273001	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:	

On **Page 2**, set the **Skill** field to **y** as shown below.

add hunt-group 33		Page 2 of 4
HUNT GROUP		
Skill? y	Expected Call Handling Time (sec): 180	
AAS? n		
Measured: none		
Supervisor Extension:		
Controlling Adjunct: none		
Timed ACW Interval (sec):		
Multiple Call Handling: none		

Repeat the above steps to create a hunt groups for the outbound service, hunt group **34** is shown below.

add hunt-group 34		Page 1 of 4
HUNT GROUP		
Group Number: 34	ACD? y	
Group Name: Presenceco Outbound	Queue? y	
Group Extension: 8273002	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:	

On **Page 2**, set the **Skill** field to **y** as shown below.

add hunt-group 34		Page 2 of 4
HUNT GROUP		
Skill? y	Expected Call Handling Time (sec): 180	
AAS? n		
Measured: none		
Supervisor Extension:		
Controlling Adjunct: none		
Timed ACW Interval (sec):		
Multiple Call Handling: none		

## 5.4.2 Vectors

Enter the **change vector n** command, where **n** is the vector number. The adjunct routing link enables Presenceco Presence Server to specify the destination of a call. The **adjunct routing link** number is defined by the position of the AESVCS link on **Page 3** of the ip-services (not shown), in this case Server ID **1**. Enter the vector steps to queue to **skill 33** as shown below. Skill 33 relates to the skill enabled hunt group configured previously.

change vector 3				Page 1 of 6	
CALL VECTOR					
Number: 3		Name: Presenceco In			
Multimedia? n	Attendant Vectoring? n	Meet-me Conf? n	Lock? n		
Basic? y	EAS? y	G3V4 Enhanced? y	ANI/II-Digits? y	ASAI Routing? y	
Prompting? y	LAI? y	G3V4 Adv Route? y	CINFO? y	BSR? y	Holidays? y
Variables? y	3.0 Enhanced? y				
01 adjunct	routing link 1				
02 wait-time	5 secs hearing silence				
03 queue-to	skill 33 pri m				
04 wait-time	999 secs hearing ringback				

The above step may also be used to create a Vector for the Outbound service, as shown below.

change vector 4				Page 1 of 6			
CALL VECTOR							
Number: 4		Name: Presenceco Outb					
Multimedia? n	Attendant Vectoring? n		Meet-me Conf? n		Lock? n		
Basic? y	EAS? y	G3V4 Enhanced? y	ANI/II-Digits? y		ASAI Routing? y		
Prompting? y	LAI? y	G3V4 Adv Route? y	CINFO? y		BSR? y	Holidays? y	
Variables? y	3.0 Enhanced? y						
01 adjunct	routing link 1						
02 wait-time	5 secs hearing silence						
03 queue-to	skill 34 pri m						
04 wait-time	999 secs hearing ringback						

## 5.4.3 Vector Directory Numbers (VDN)

Enter the **add vdn n** command, where **n** is an available extension number. On **Page 1** assign a **Name** for the VDN and set the **Vector Number** to the relevant vector.

add vdn 8274001		Page 1 of 3	
VECTOR DIRECTORY NUMBER			
Extension: 8274001			
Name*: Presenceco Inbound			
Destination: Vector Number		3	
Attendant Vectoring? n			
Meet-me Conferencing? n			
Allow VDN Override? n			
COR: 1			
TN*: 1			
Measured: none			
VDN of Origin Annc. Extension*:			
1st Skill*:			
2nd Skill*:			
3rd Skill*:			

The above step may also be used to create a VDN for the Outbound service, as shown below.

```
add vdn 8274002                                     Page 1 of 3
                                         VECTOR DIRECTORY NUMBER

      Extension: 8274002
      Name*: Presenceco Outbound
      Destination: Vector Number 4
      Attendant Vectoring? n
      Meet-me Conferencing? n
      Allow VDN Override? n
      COR: 1
      TN*: 1
      Measured: none

      VDN of Origin Annc. Extension*:
      1st Skill*:
      2nd Skill*:
      3rd Skill*:
```

## 5.5 Administer Class of Restriction

Enter the **change cor 1** command where **1** corresponds to the Class of Restriction assigned to the trunk group in **Section 5.3** and the agent login IDs in **Section 5.6**. On **Page 1**, set the **Direct Agent Calling** to **y**. This will allow agents to be called directly once they are logged in.

```
change cor 1                                         Page 1 of 23
                                         CLASS OF RESTRICTION

      COR Number: 1
      COR Description: Default

      FRL: 0                                         APLT? y
      Can Be Service Observed? y                   Calling Party Restriction: none
      Can Be A Service Observer? y                 Called Party Restriction: none
      Time of Day Chart: 1                         Forced Entry of Account Codes? n
      Priority Queuing? n                          Direct Agent Calling? y
      Restriction Override: all                    Facility Access Trunk Test? n
      Restricted Call List? n                      Can Change Coverage? n n
```

## 5.6 Administer Agent Logins

Enter the **add agent-loginID n** command; where **n** is an available extension number. Enter a descriptive name for the agent in the **Name** field. Ensure the **COR** field is set to **1** which relates to the COR configured in **Section 5.5**. The **Auto Answer** field is set to **station** except for those logins that will be used for outbound services. In that case, the field will be set to **all**. Configure a password as required.

add agent-loginID 8271001		Page 1 of 3	
AGENT LOGINID			
Login ID: 8271001		AAS? n	
Name: Presenceco Agent 1		AUDIX? n	
TN: 1		LWC Reception: spe	
COR: 1		LWC Log External Calls? n	
Coverage Path:		AUDIX Name for Messaging:	
Security Code:		LoginID for ISDN/SIP Display? n	
		Password:	
		Password (enter again):	
		Auto Answer: station	
		MIA Across Skills: system	

On **Page 2**, assign a skill to the agent by entering the relevant hunt group number created in **Section 5.4.1** for **SN** and entering a skill level of **1** for **SL**. In this case, an agent is able to handle both inbound and outbound calls is created. Set the **Direct Agent Skill** to the Inbound hunt group **33**.

change agent-loginID 201		Page 2 of 3	
AGENT LOGINID			
Direct Agent Skill: 33		Service Objective? n	
Call Handling Preference: skill-level		Local Call Preference? n	
SN	RL SL	SN	RL SL
1: 33	1	16:	
2: 34	1	17:	

Repeat this task accordingly for any additional inbound or outbound agents required.

## 5.7 Configure Agent Stations

For each station that agents will log in to, enter the command **change station n**, where **n** is the station extension. On **Page 4**, the following buttons must be assigned as shown below:

- **aux-work** – Agent is logged in to the ACD but is not available to take a call.
- **manual-in** – Agent is available to accept ACD calls.
- **after-call** – Agent state after the ACD call is completed. The agent is not available.
- **release** – State when the call is dropped.

<b>change station 8270001</b>		<b>Page 4 of 5</b>	
STATION			
SITE DATA			
Room:		Headset?	n
Jack:		Speaker?	n
Cable:		Mounting:	d
Floor:		Cord Length:	0
Building:		Set Color:	
ABBREVIATED DIALING			
List1:	List2:	List3:	
BUTTON ASSIGNMENTS			
1: call-appr	5: manual-in	Grp:	
2: call-appr	6: after-call	Grp:	
3: call-appr	7: release		
4: aux-work	8::		
RC:	Grp:		

## 5.8 Administer Phantom Stations

Presence Suite uses stations via AES to initiate calls on Communication Manager. These stations will be used to place calls to customers for outbound services as well as to place calls to agents in order to reserve an agent to handle the outbound call. Use the command **add station n**, enter a descriptive name for **Name** and enter **X** for the **Port**. Extensions **8270400** to **8270402** were created.

<b>add station 8270400</b>		<b>Page 1 of 5</b>	
STATION			
Extension: 8270400	Lock Messages? n	BCC: 0	
Type: 6408D+	Security Code:	TN: 1	
Port: X	Coverage Path 1:	COR: 1	
Name: Presenceco Phantom	Coverage Path 2:	COS: 1	
	Hunt-to Station:		
STATION OPTIONS			
Loss Group: 1		Time of Day Lock Table:	
Data Module? n	Personalized Ringing Pattern: 1		
Display Module? n	Message Lamp Ext: 3500		
Survivable COR: internal	Media Complex Ext:		
Survivable Trunk Dest? y			



## 5.9 Note procr IP Address for Avaya Aura® Application Enablement Services Connectivity

Display the procr IP Address by using the command **display node-names ip** and noting the IP address for the **procr** and AES (**aes7vm**).

display node-names ip		Page 1 of 2
		IP NODE NAMES
Name	IP Address	
SM100	10.10.40.34	
<b>AES71678</b>	<b>10.10.40.30</b>	
default	0.0.0.0	
g430	10.10.40.15	
<b>procr</b>	<b>10.10.40.31</b>	

## 5.10 Configure Transport Link for Avaya Aura® Application Enablement Services Connectivity

To administer the transport link to AES use the **change ip-services** command. On **Page 1** add an entry with the following values:

- **Service Type:** should be set to **AESVCS**.
- **Enabled:** set to **y**.
- **Local Node:** set to the node name assigned for the procr in **Section 5.9**.
- **Local Port** Retain the default value of **8765**.

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

Go to **Page 4** of the **ip-services** form and enter the following values:

- **AE Services Server:** Name obtained from the AES server, in this case **AES71678**.
- **Password:** Enter a password to be administered on the AES server.
- **Enabled:** Set to **y**.

**Note:** The password entered for **Password** field must match the password on the AES server in **Section 6.2**. The **AE Services Server** should match the administered name for the AES server, this is created as part of the AES installation, and can be obtained from the AES server by typing **uname -n** at the Linux command prompt.

change ip-services				Page 4 of 4
AE Services Administration				
Server ID	AE Services Server	Password	Enabled	Status
1:	AES71678	*****	y	idle
2:				
3:				

## 5.11 Configure CTI Link for TSAPI Service

Add a CTI link using the **add cti-link n** command. Enter an available extension number in the **Extension** field. 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	Page	1 of 3
CTI LINK			
CTI Link: 1			
Extension: 2002			
Type: ADJ-IP			
		COR: 1	
Name: aes7vm			

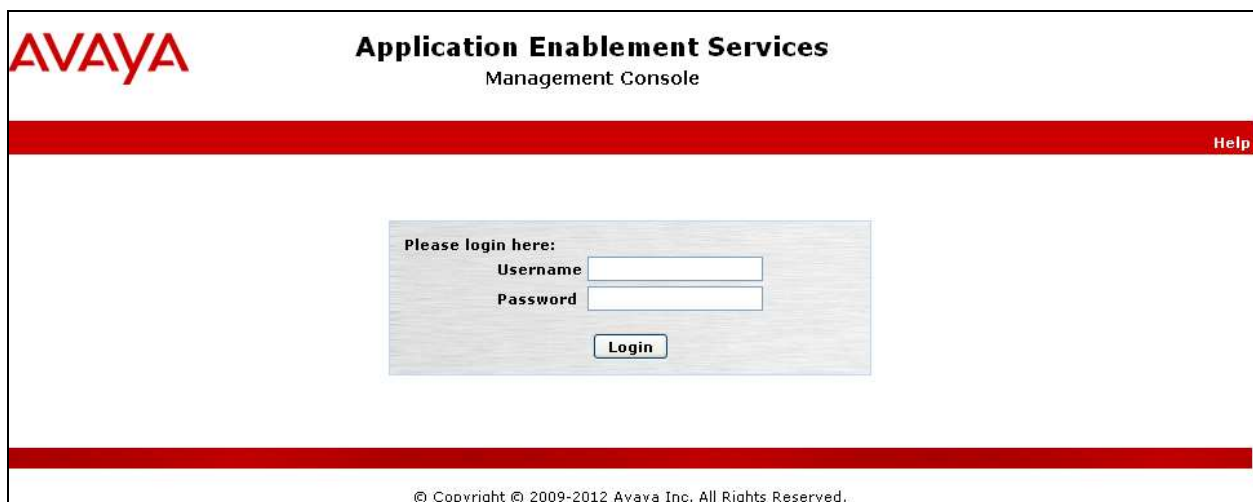
## 6. Configure Avaya Aura® Application Enablement Services Server

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

- Verify Licensing
- Create Switch Connection
- Administer TSAPI link
- Create CTI User
- Enable CTI Link User
- Identify Tlinks

### 6.1 Verify Licensing

To access the maintenance console, enter **https://<ip-addr>** as the URL in an Internet browser, where <ip-addr> is the active IP address of AES. The login screen is displayed, log in with the appropriate credentials and then select the **Login** button.



The screenshot shows the Avaya Application Enablement Services Management Console login interface. At the top left is the Avaya logo. To its right, the text "Application Enablement Services" is displayed in bold, with "Management Console" underneath it. A red horizontal bar spans the width of the page, with the word "Help" in white text on the right side. In the center of the page is a login box with a light gray background. Inside this box, the text "Please login here:" is followed by two input fields: "Username" and "Password". Below these fields is a "Login" button. At the bottom of the page, a red horizontal bar is present, and below it, the copyright notice "© Copyright © 2009-2012 Avaya Inc. All Rights Reserved." is displayed.

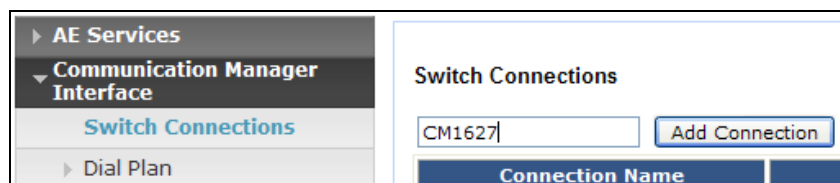
The Application Enablement Services Management Console appears displaying the **Welcome to OAM** screen (not shown). Select **AE Services** and verify that the TSAPI Service is licensed by ensuring that **TSAPI Service** is in the list of services and that the **License Mode** is showing **NORMAL MODE**. If not, contact an Avaya support representative to acquire the proper license for your solution.



Service	Status	State	License Mode
ASAI Link Manager	N/A	Running	N/A
CVLAN Service	ONLINE	Running	NORMAL MODE
DLG Service	OFFLINE	Running	N/A
DMCC Service	ONLINE	Running	NORMAL MODE
<b>TSAPI Service</b>	ONLINE	Running	<b>NORMAL MODE</b>
Transport Layer Service	N/A	Running	N/A

## 6.2 Create Switch Connection

From the AES Management Console navigate to **Communication Manager Interface** → **Switch Connections** to set up a switch connection. Enter in a name for the Switch Connection to be added and click the **Add Connection** button.

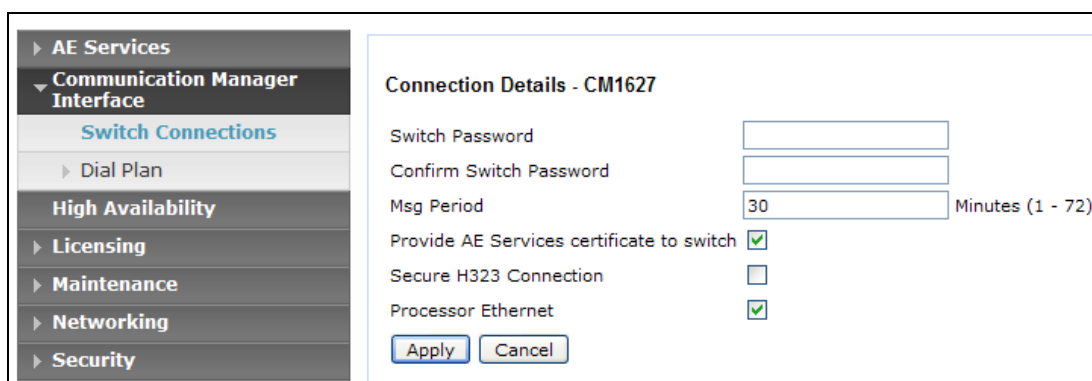


Switch Connections

CM1627

Connection Name

In the resulting screen enter the **Switch Password**, the Switch Password must be the same as that entered into Communication Manager AE Services Administration screen via the **change ip-services** command, described in **Section 5.10**. Default values may be accepted for the remaining fields. Click **Apply** to save changes.



Connection Details - CM1627

Switch Password

Confirm Switch Password

Msg Period  Minutes (1 - 72)

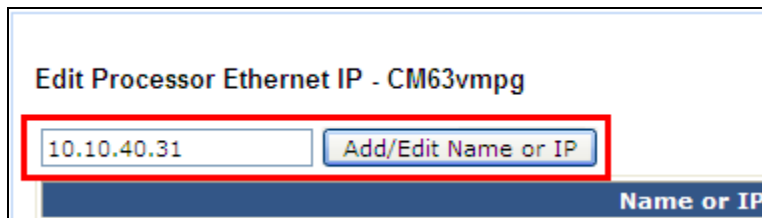
Provide AE Services certificate to switch ☒

Secure H323 Connection ☐

Processor Ethernet ☒

From the **Switch Connections** screen, select the radio button for the recently added switch connection and select the **Edit CLAN IPs** button (not shown). In the resulting screen, enter the

IP address of the **procr** as shown in **Section 5.9** that will be used for the AES connection and select the **Add Name or IP** button.



Edit Processor Ethernet IP - CM63vmpg

10.10.40.31 Add/Edit Name or IP

Name or IP

### 6.3 Administer TSAPI link

From the Application Enablement Services Management Console, select **AE Services** → **TSAPI** → **TSAPI Links**. Select **Add Link** button as shown in the screen below.



TSAPI Links

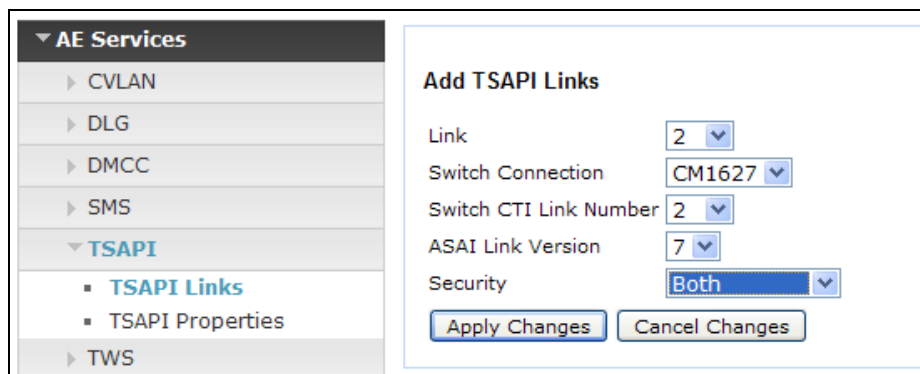
Link	Switch Connection

Add Link Edit Link Delete Link

On the **Add TSAPI Links** screen, enter the following values:

- **Link:** Use the drop-down list to select an unused link number.
- **Switch Connection:** Choose the switch connection **CM1627**, which has already been configured in **Section 6.2**, from the drop-down list.
- **Switch CTI Link Number:** Corresponding CTI link number configured in **Section 5.11**.
- **ASAI Link Version:** This can be left at the default value of **7**.
- **Security:** This can be left at the default value. The value **both** was used in this test.

Once completed, select **Apply Changes**.



▼ AE Services

- ▶ CVLAN
- ▶ DLG
- ▶ DMCC
- ▶ SMS
- ▼ TSAPI
  - TSAPI Links
  - TSAPI Properties
- ▶ TWS

**Add TSAPI Links**

Link 2 ▼

Switch Connection CM1627 ▼

Switch CTI Link Number 2 ▼

ASAI Link Version 7 ▼

Security Both ▼

Apply Changes Cancel Changes

Another screen appears for confirmation of the changes. Choose **Apply**.

**Apply Changes to Link**  

Warning! Are you sure you want to apply the changes?  
These changes can only take effect when the TSAPI server restarts.  
Please use the Maintenance -> Service Controller page to restart the TSAPI server.

The TSAPI Service must be restarted to effect the changes made in this section. From the Management Console menu, navigate to **Maintenance** → **Service Controller**. On the Service Controller screen, tick the **TSAPI Service** and select **Restart Service**.

Communication Manager Interface

Licensing

Maintenance

Date Time/NTP Server

Security Database

Service Controller

Server Data

Networking

Security

Status

Service Controller

Service	Controller Status
<input type="checkbox"/> ASAI Link Manager	Running
<input type="checkbox"/> DMCC Service	Running
<input type="checkbox"/> CVLAN Service	Running
<input type="checkbox"/> DLG Service	Running
<input type="checkbox"/> Transport Layer Service	Running
<input checked="" type="checkbox"/> TSAPI Service	Running

For status on actual services, please use [Status and Control](#)

## 6.4 Create Avaya CTI User

A User ID and password needs to be configured for the Presence Suite server to communicate as a TSAPI client with the Application Enablement Services server. Navigate to the **User Management** → **User Admin** screen then choose the **Add User** option (not shown). In the **Add User** screen shown below, enter the following values:

- **User Id** - This will be used by the Presence Suite Server in **Section 7.1**.
- **Common Name** and **Surname** - Descriptive names need to be entered.
- **User Password** and **Confirm Password** - This will be used with the **User Id** in **Section 7.1**.
- **CT User** - Select **Yes** from the drop-down menu.

Complete the process by choosing **Apply** at the bottom of the screen (not shown).

The screenshot shows the 'User Management | User Admin | List All Users' interface. The left sidebar contains a navigation menu with the following items: AE Services, Communication Manager Interface, Licensing, Maintenance, Networking, Security, Status, User Management (expanded), Service Admin, User Admin (expanded), Add User, Change User Password, List All Users (selected), Modify Default Users, Search Users, Utilities, and Help. The main content area is titled 'Edit User' and contains the following fields: \* User Id (presence), \* Common Name (presence), \* Surname (presence), User Password, Confirm Password, Admin Note, Avaya Role (None), Business Category, Car License, CM Home, Cms Home, CT User (Yes), Department Number, Display Name, Employee Number, Employee Type, and Extension Number. Two red boxes highlight the fields for \* User Id, \* Common Name, and \* Surname, and the CT User dropdown menu.

The next screen will show a message indicating that the user was created successfully (not shown).

## 6.5 Enable Unrestricted Access for CTI User

Navigate to the **CTI Users** screen by selecting **Security** → **Security Database** → **CTI Users** → **List All Users**. Select the user that was created in **Section 6.4** and select the **Edit** option (not shown). The **Edit CTI User** screen appears. Check the **Unrestricted Access** box and **Apply Changes** at the bottom of the screen.

The screenshot displays the 'Edit CTI User' interface. On the left, a sidebar menu lists various system components, with 'Security' expanded to show 'Security Database' and 'CTI Users'. The main content area is titled 'Edit CTI User' and contains the following fields and controls:

- User Profile:**
  - User ID: ctiuser
  - Common Name: ctiuser
  - Worktop Name: NONE (dropdown)
  - Unrestricted Access:** ☒ (highlighted with a red box)
- Call and Device Control:**
  - Call Origination/Termination and Device Status: None (dropdown)
- Call and Device Monitoring:**
  - Device Monitoring: None (dropdown)
  - Calls On A Device Monitoring: None (dropdown)
  - Call Monitoring: ☐
- Routing Control:**
  - Allow Routing on Listed Devices: None (dropdown)

At the bottom of the form, there are two buttons: **Apply Changes** (highlighted with a red box) and **Cancel Changes**.

A screen (not shown) appears to confirm applied changes to CTI User, choose **Apply**. This CTI user should now be enabled.



## 6.6 Identify Tlinks

Navigate to **Security** → **Security Database** → **Tlinks**. Verify the value of the **Tlink Name**. This will be needed to configure Presence Suite in **Section 7.1**.

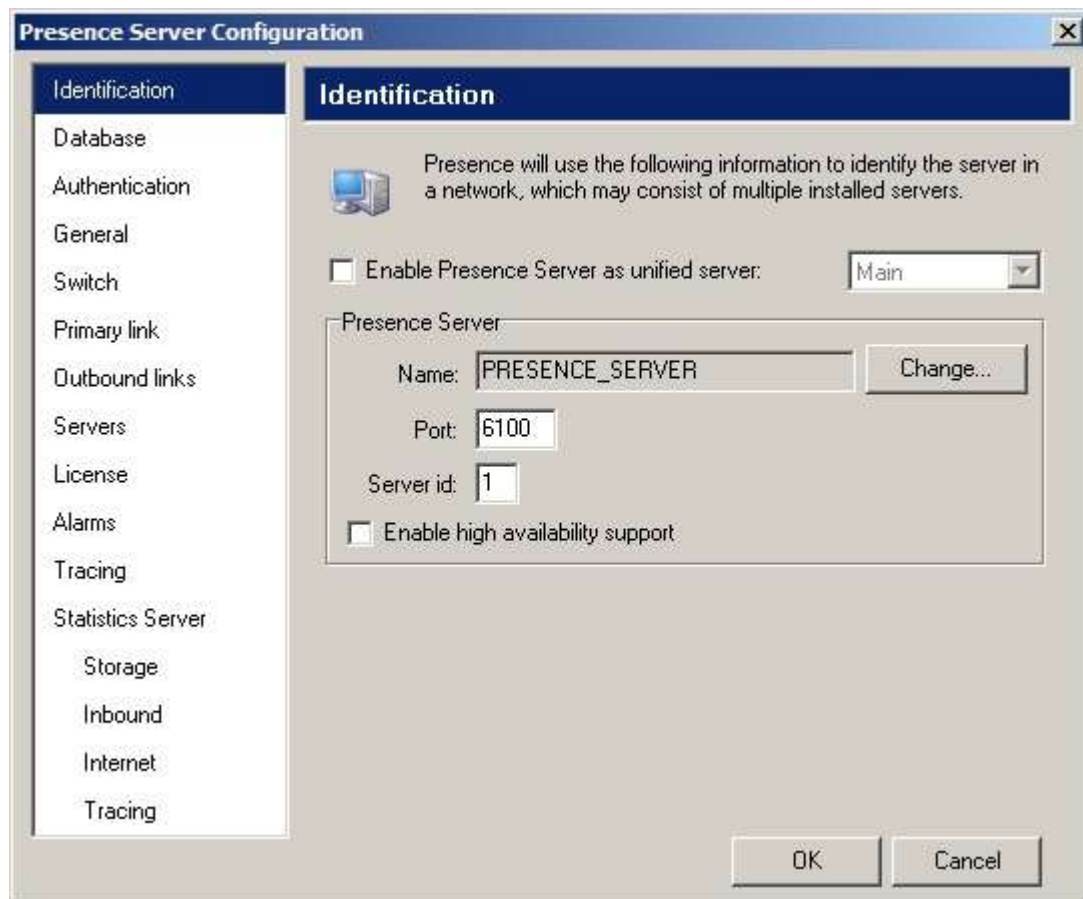
The screenshot displays the Avaya Management System (AMS) interface. On the left is a navigation tree with the following items: AE Services, Communication Manager Interface, High Availability, Licensing, Maintenance, Networking, Security (expanded), Account Management, Audit, Certificate Management, Enterprise Directory, Host AA, PAM, Security Database (expanded), Control, CTI Users, Devices, Device Groups, and Tlinks (highlighted in blue). The main content area on the right is titled 'Tlinks'. It contains a 'Tlink Name' section with two radio button options: 'AVAYA#CM1627#CSTA#AES71678' (which is selected) and 'AVAYA#CM1627#CSTA-S#AES71678'. Below these options is a 'Delete Tlink' button.

## 7. Configure the Presence Suite Server

The Presence Suite includes the Presence Server, Presence Mail Interactions Server, Presence Web Interactions Server, Presence Administrator, Presence Supervisor, and Presence Agent. The Presence Server and the Oracle database were pre-installed on the same machine for convenience during the compliance testing. The Presence server was configured and provided by Presence Technology. An outline of the configuration relevant to the Avaya solution integration is detailed below.

### 7.1 Presence Server Configuration

Launch the Presence Server configuration application by double clicking the **pcoSERVERCFG.exe** located in the pre-installed Presence folder on the Presence Server (not shown). Select the **Identification** option from the menu on the left side of the screen, enter the **Server name** as **PRESENCE\_SERVER** as used for the identification of the server. The **Port** can be set to **6100**. Note that the actual value for server port can vary. Press **OK** to continue.



The screenshot shows the 'Presence Server Configuration' window with the 'Identification' tab selected. The left sidebar lists various configuration categories: Identification, Database, Authentication, General, Switch, Primary link, Outbound links, Servers, License, Alarms, Tracing, Statistics Server, Storage, Inbound, Internet, and Tracing. The main area of the window is titled 'Identification' and contains the following information:

- A computer icon and text: 'Presence will use the following information to identify the server in a network, which may consist of multiple installed servers.'
- A checkbox labeled 'Enable Presence Server as unified server:' with a dropdown menu set to 'Main'.
- A section titled 'Presence Server' containing:
  - A text field for 'Name:' with the value 'PRESENCE\_SERVER' and a 'Change...' button.
  - A text field for 'Port:' with the value '6100'.
  - A text field for 'Server id:' with the value '1'.
  - A checkbox labeled 'Enable high availability support'.
- 'OK' and 'Cancel' buttons at the bottom right.

Select **General** from the menu on the left side of the screen. If desired, the Maintenance configuration values can be altered here, for the compliance test the default values were retained.

The screenshot shows the 'Presence Server Configuration' dialog box with the 'General' tab selected. The left sidebar lists various configuration categories, with 'General' highlighted. The main area contains two sections: 'Maintenance configuration values' and 'Other'. The 'Maintenance configuration values' section includes four settings: 'Check for pending outbound calls every' (30 seconds), 'Minimum time between queue updates in server (in minutes)' (15), 'Time for reorganizing queues in server' (03:00), and 'Keep server events from last' (15 days). The 'Other' section includes 'Length of area codes' (6 digits). At the bottom right are 'OK' and 'Cancel' buttons.

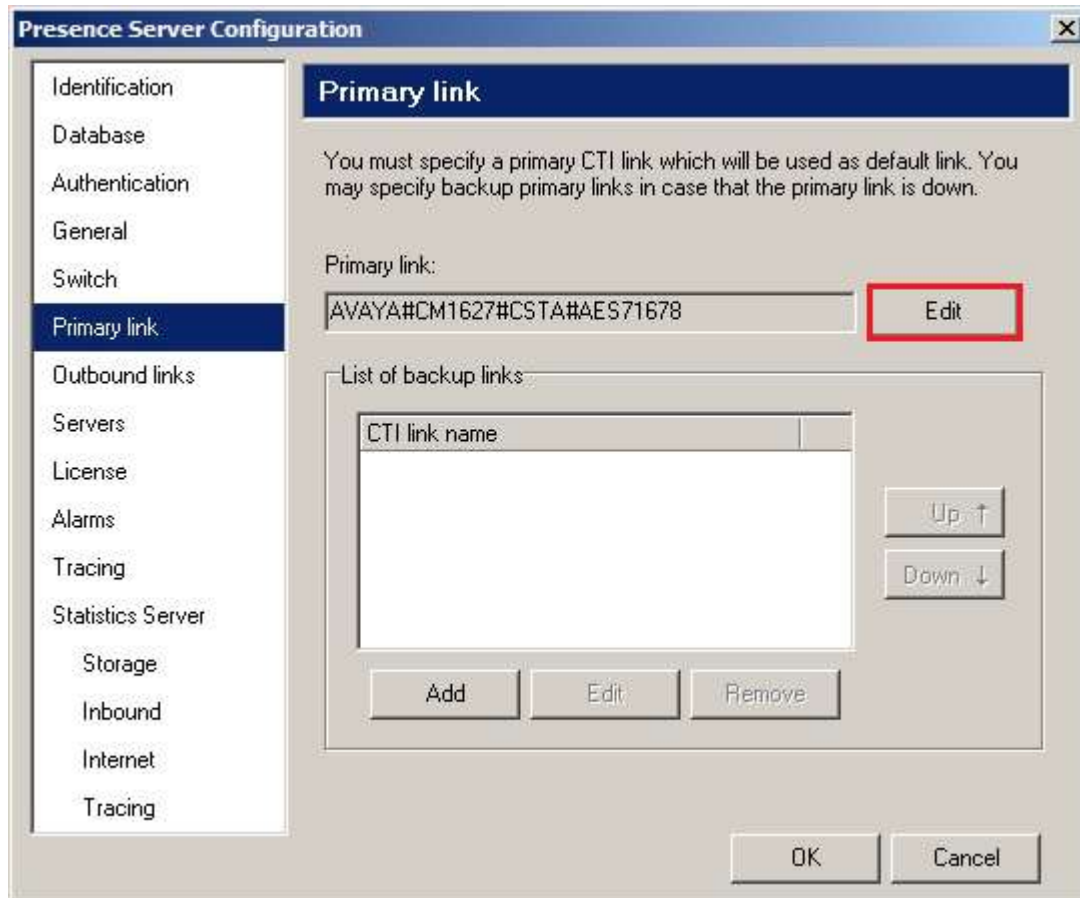
Configuration Category	Setting	Value	Unit
Maintenance configuration values	Check for pending outbound calls every	30	seconds
	Minimum time between queue updates in server (in minutes). If a queue is updated within a shorter interval, a warning will be triggered in server:	15	minutes
	Time for reorganizing queues in server. This is a critical process which may affect the server performance:	03:00	minutes
	Keep server events from last	15	days
Other	Length of area codes:	6	digits

Select the **Switch** option from the menu on the left side of the screen. The **System login to be assigned to contacts not handled by an agent (CTI login)** field should be set to a value supplied by Presence, the value used for this configuration is **99999**. Check the **Specify phantom extension for preview mode** checkbox and enter the phantom extensions configured in **Section 5.9**.

The screenshot shows the 'Presence Server Configuration' dialog box with the 'Switch' tab selected. The left sidebar contains a list of configuration categories: Identification, Database, Authentication, General, Switch (highlighted), Primary link, Outbound links, Servers, License, Alarms, Tracing, Statistics Server, Storage, Inbound, Internet, and Tracing. The main area of the dialog is divided into three sections. The top section, titled 'Switch', contains two text input fields: 'Prefix for outgoing calls:' (empty) and 'System login to be assigned to contacts not handled by an agent (CTI login):' (containing '99999'). The middle section, titled 'Preview mode', contains a checked checkbox labeled 'Specify phantom extensions' and a text input field containing '8270400-8270402'. The bottom section, titled 'Presence IVR', contains two text input fields: 'Port for integration:' (containing '9090') and 'Service extension (VDN) for routing:' (empty). At the bottom right of the dialog are 'OK' and 'Cancel' buttons.

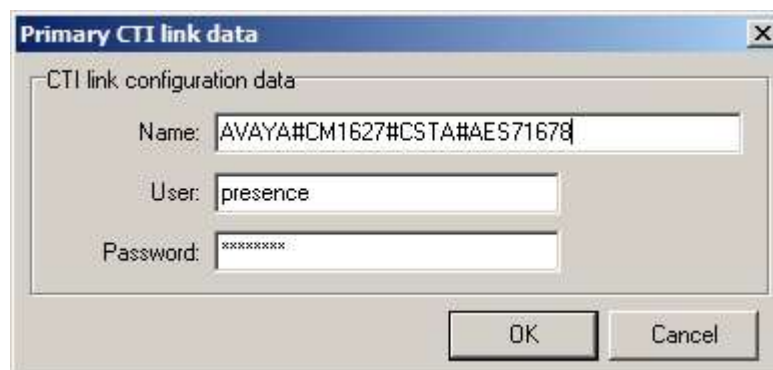
Field	Value
Prefix for outgoing calls:	
System login to be assigned to contacts not handled by an agent (CTI login):	99999
Specify phantom extensions (checkbox)	Checked
Phantom extensions text field	8270400-8270402
Port for integration:	9090
Service extension (VDN) for routing:	

Select the **Primary link** menu on the left side of the screen and choose the **Edit** button to enter a value.



The image shows the 'Presence Server Configuration' dialog box. On the left is a vertical menu with options: Identification, Database, Authentication, General, Switch, **Primary link** (highlighted), Outbound links, Servers, License, Alarms, Tracing, Statistics Server, Storage, Inbound, Internet, and Tracing. The main area is titled 'Primary link' and contains the text: 'You must specify a primary CTI link, which will be used as default link. You may specify backup primary links in case that the primary link is down.' Below this, there is a 'Primary link:' label followed by a text field containing 'AVAYA#CM1627#CSTA#AES71678'. To the right of this field is an 'Edit' button, which is highlighted with a red rectangle. Below the primary link field is a section titled 'List of backup links' containing a list box with the header 'CTI link name'. To the right of the list box are 'Up ↑' and 'Down ↓' buttons. Below the list box are 'Add', 'Edit', and 'Remove' buttons. At the bottom right of the dialog are 'OK' and 'Cancel' buttons.

In the resulting pop-up box enter the Tlink name from **Section 6.6** in the **Name** field. For the **User** and **Password** fields enter the user name and password configured on the Application Enablement Services in **Section 6.4**. Click **OK**.



The image shows the 'Primary CTI link data' dialog box. It has a title bar with a close button. The main area is titled 'CTI link configuration data' and contains three fields: 'Name' with the value 'AVAYA#CM1627#CSTA#AES71678', 'User' with the value 'presence', and 'Password' with a masked value 'xxxxxxxx'. At the bottom right are 'OK' and 'Cancel' buttons.

## 7.2 Presence Service Configuration

A number of services for inbound, outbound, email and internet were configured via the Presence Administrator. This section covers the basic configuration for each type of service. Please refer to **Section 10** for detailed documentation on configuring Presence Suite services.

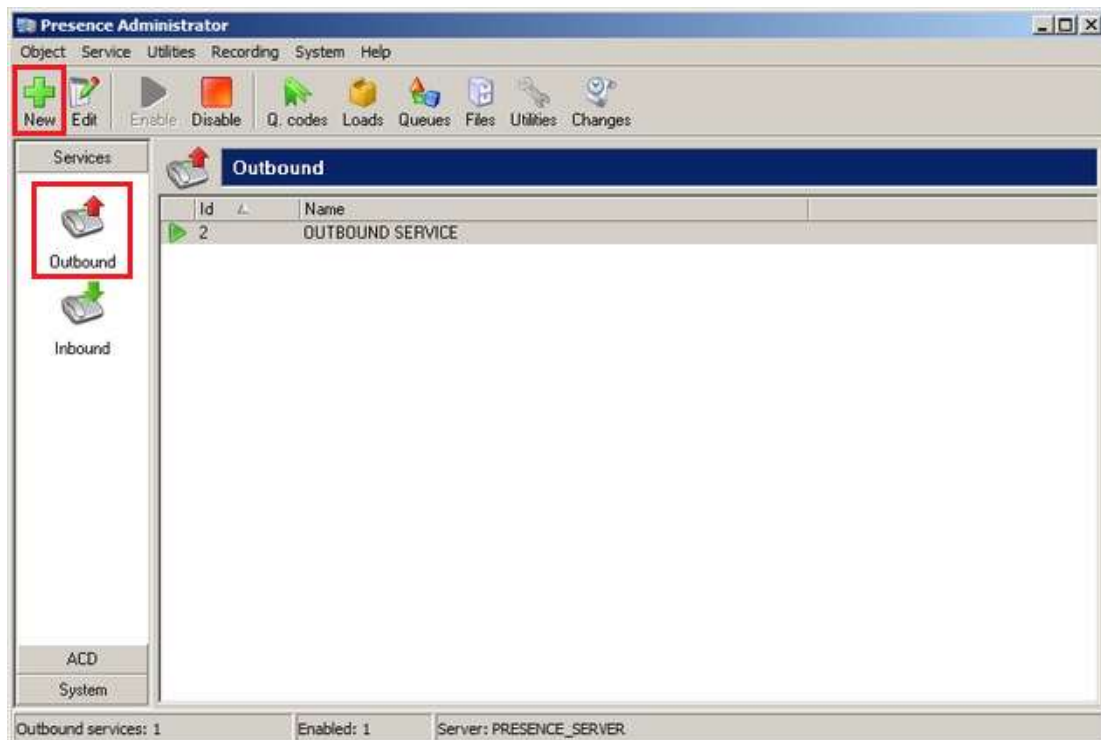
### 7.2.1 Logging in to Presence Administrator

Launch the **Presence Administrator** application by double clicking the **pcadmin.exe** located in the Presence folder (not shown). The username and password that appear in the **User** and **Password** fields are created during the Presence Server installation.



### 7.2.2 Outbound Service

After logging in to Presence Administrator the following screen will be displayed. Select **Services** → **Outbound** from the Presence Administrator main menu on the left hand side. Click the **New** button to configure an outbound service.



In the resulting screen, select **General** from the menu on the left hand side and enter a **Name** for the outbound service. In the **Calling hours** field set the time range for which the outbound service will be active. All other fields are left with their default values. Click **OK** to save service.

The screenshot shows the 'Outbound service' configuration window with the 'General' tab selected. The left sidebar lists various configuration options, with 'General' at the top. The main area contains the following fields and controls:

- Id:** 2
- Name:** OUTBOUND SERVICE (highlighted with a red box)
- Resource profile:** General (dropdown menu)
- Stop reasons:** [All] (dropdown menu)
- Scheduled calling hours:**
  - Do not schedule records for the last 15 minutes of a time range (dropdown menu)
  - ☐ Limit date: 16/12/2015 (dropdown menu)
- Outbound calling hours:** 08:00-22:00 (highlighted with a red box)

At the bottom right, there are 'OK' and 'Cancel' buttons.



Select **Outbound type** from the left hand side menu and moving to the right, select the **Type** of outbound service. This specifies the mode in which the outbound service will operate, for further details of the type of outbound service available please refer to documentation in **Section 10**. In the **Extension/Skill** field enter the extension number assigned to the outbound hunt group configured in **Section 5.4.1**. In the **VDN/CDN** field enter the VDN number assigned to Outbound calls configured in **Section 5.4.3**. In the test configuration only one CTI link was configured so the **CTI Link** field is set to <<**Primary CTI Link**>> if multiple CTI links exist on the system then the specific CTI link can be specified. All other field may be left at their default values.

**Outbound service**

**Outbound type**

Type: Progressive

ACD Items

Extension/Skill: 8273002

VDN/CDN: 8274002

CTI link: <<Primary CTI link>>

☒ Use primary CTI link in case that CTI link is not connected

☐ Maximum number of concurrent service calls:

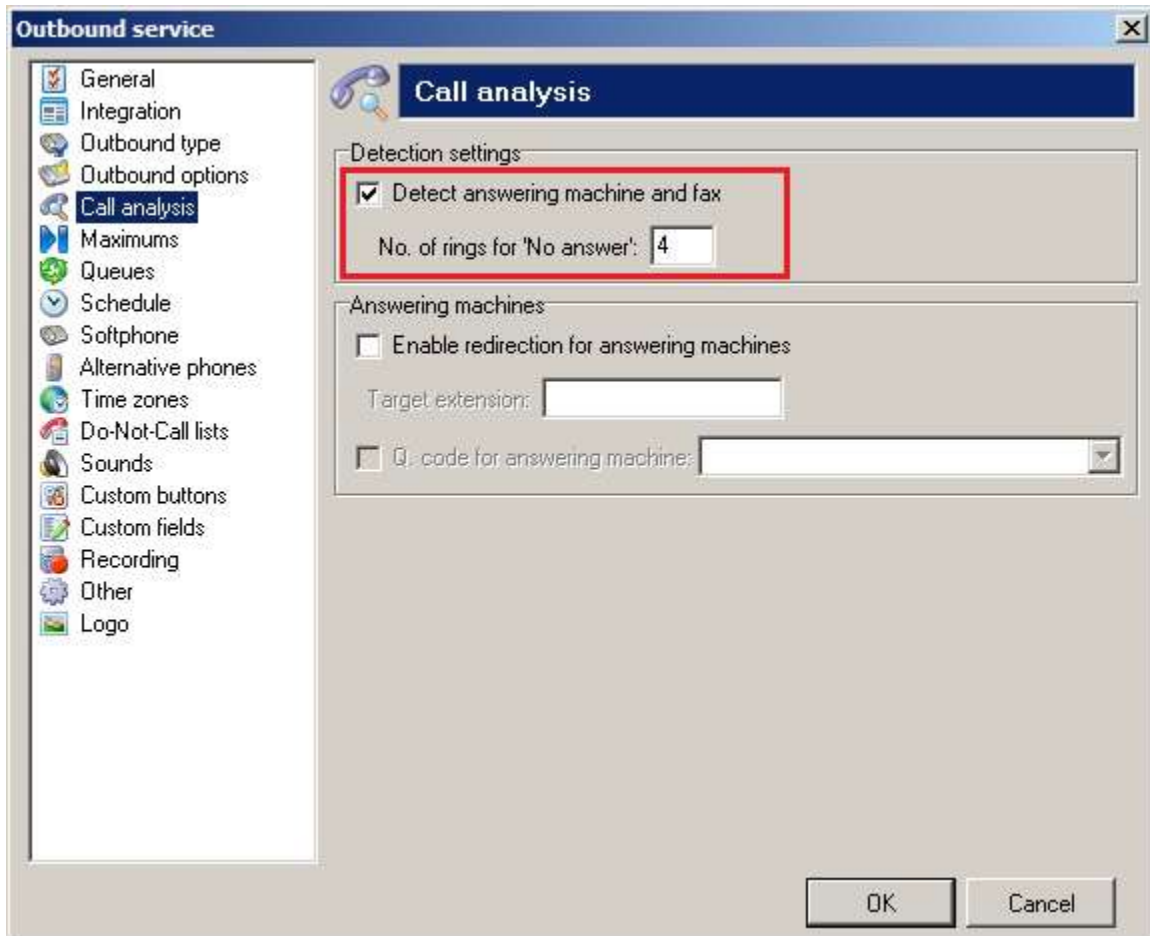
☐ Check agent availability

☐ Minimum number/percentage of available agents:

OK Cancel

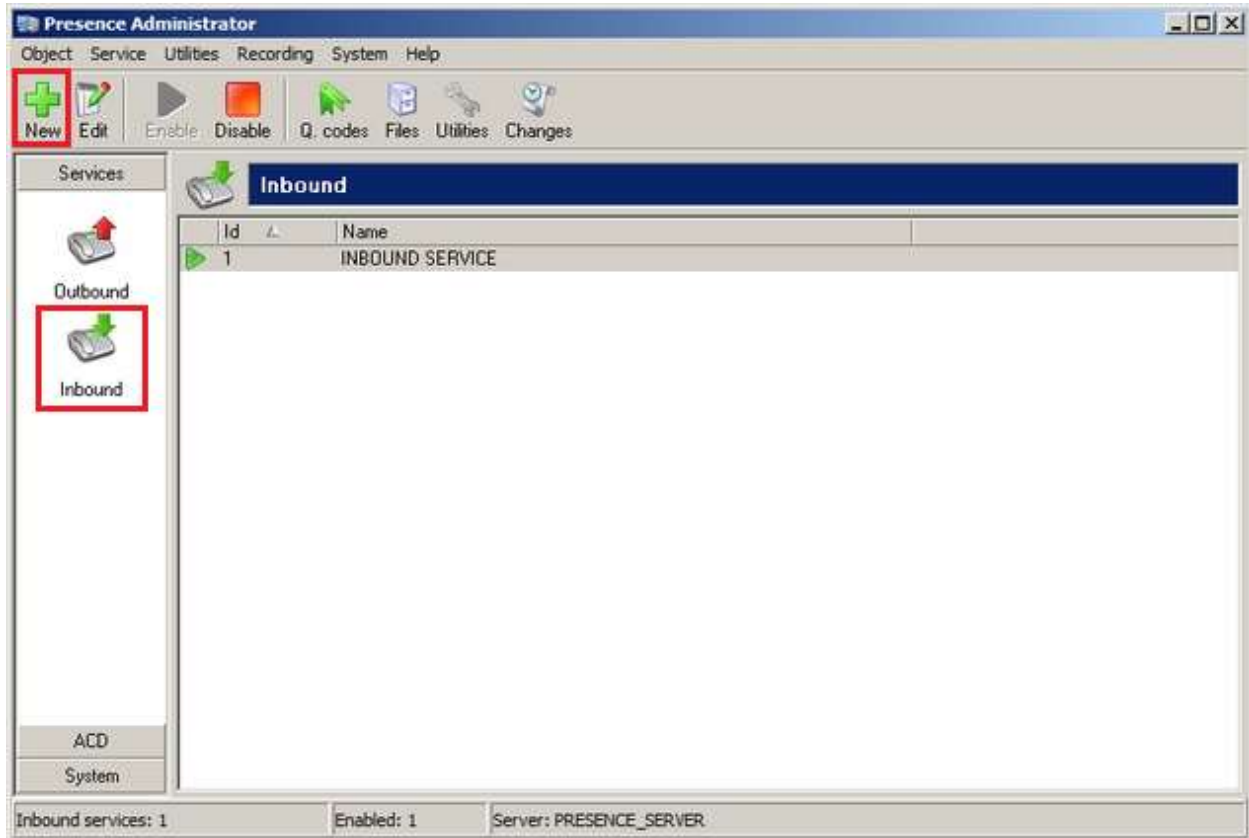


Select **Call analysis** from the left hand side menu. The fields in the right hand side define how the outbound service should behave following an unsuccessful attempt at contacting the customer. For testing, the **Detect answering machine and fax** box are checked and the **No. of rings for 'No answer'** is set to 4, as shown in the screen below. Click **OK** to complete the outbound service configuration.



### 7.2.3 Inbound Service

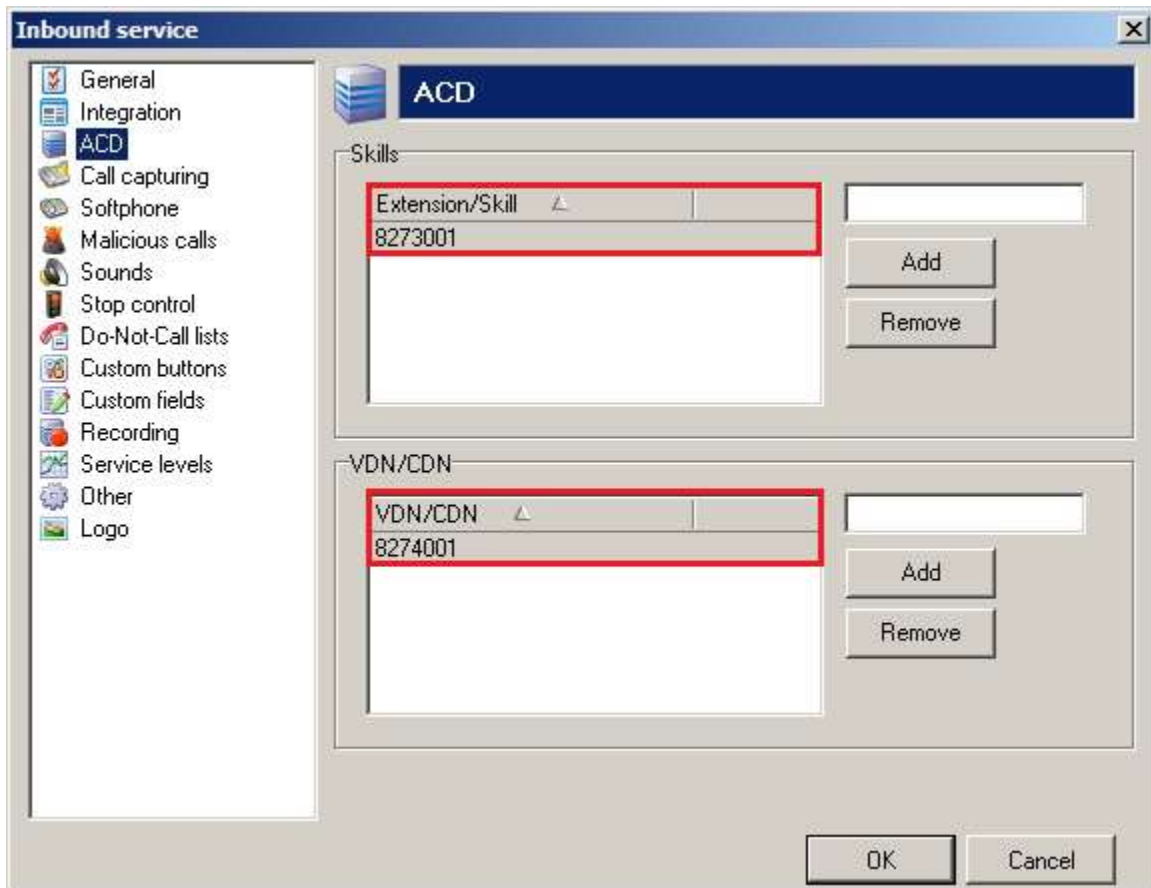
To configure an inbound service, from the left hand side select **Services** → **Inbound** from the Presence Administrator main menu. Click the **New** button.



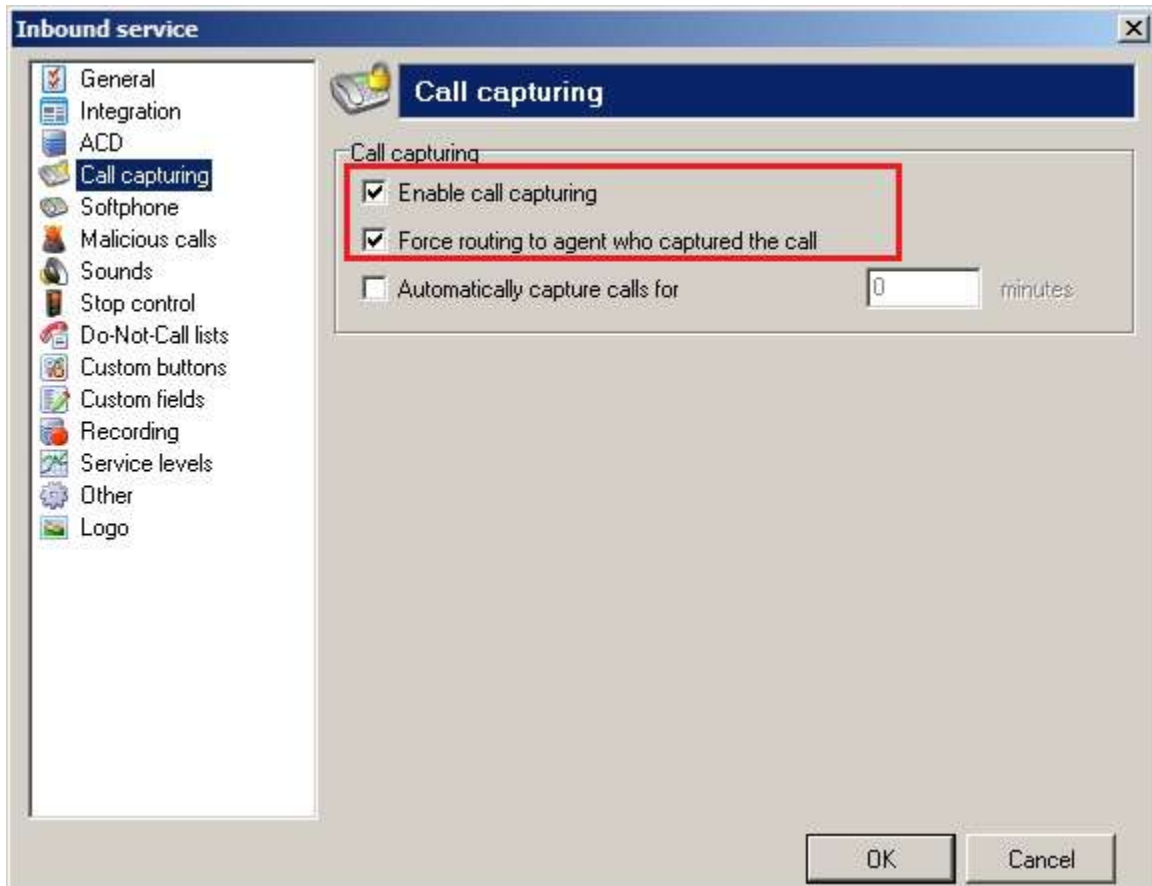
In the resulting screen, select **General** from the menu on the left hand side and enter a **Name** for the inbound service. All other fields are left with their default values.

The screenshot shows a software window titled "Inbound service". On the left is a vertical menu with icons and labels: General, Integration, ACD, Call capturing, Softphone, Malicious calls, Sounds, Stop control, Do-Not-Call lists, Custom buttons, Custom fields, Recording, Service levels, Other, and Logo. The "General" item is selected. The main area of the window has a sub-header "General" with a checkmark icon. Below this, there are four input fields: "Id:" with the value "1", "Name:" with the value "INBOUND SERVICE" (this field is highlighted with a red border), "Resource profile:" with a dropdown menu showing "General", and "Stop reasons:" with a dropdown menu showing "[All]". At the bottom right of the window are two buttons: "OK" and "Cancel".

Select **ACD** from the left hand side menu and moving to the right, under the heading **Skills** enter the skill group extensions configured in **Section 5.4.1** that will handle inbound calls in the untitled box (this includes email and web chat call types) and click **Add**. The skill group extensions will then appear to the left in the **Extension/Skill** box. Under the heading **VDN/CDN** enter the VDN configured in **Section 5.4.3** that will handle inbound calls in the untitled box and click **Add**. The VDN will then appear to the left in the **VDN/CDN** box.



Select **Call capturing** from the left hand side menu and moving to the right, select the **Enable call capturing. Force routing to agent who captured the call** was checked for this compliance testing but is each users preference. These options allow an agent to mark an inbound call so that if the caller rings back while that agent is logged onto the system, the call will be routed again to the agent who tagged the call.



Select **Malicious calls** from the left hand side menu and moving to the right, select the **Enable malicious calls detection** check box. This option allows agents to mark calls as malicious, so that the caller can be directed to another location such as a supervisor position if they call back again. In the **Target extension** field enter the extension that any malicious calls will be re-directed to. In the **VDN/CDN to control** field select the VDNs this option will be available on.

The screenshot shows the 'Inbound service' configuration window with the 'Malicious calls' tab selected. The left sidebar lists various settings, and the main area shows the configuration for malicious calls. A red box highlights the 'Enable malicious calls detection' checkbox (checked), the 'Target extension' field (8270003), and the 'VDN/CDN to control' field (checked, 8274001). The 'OK' and 'Cancel' buttons are at the bottom right.

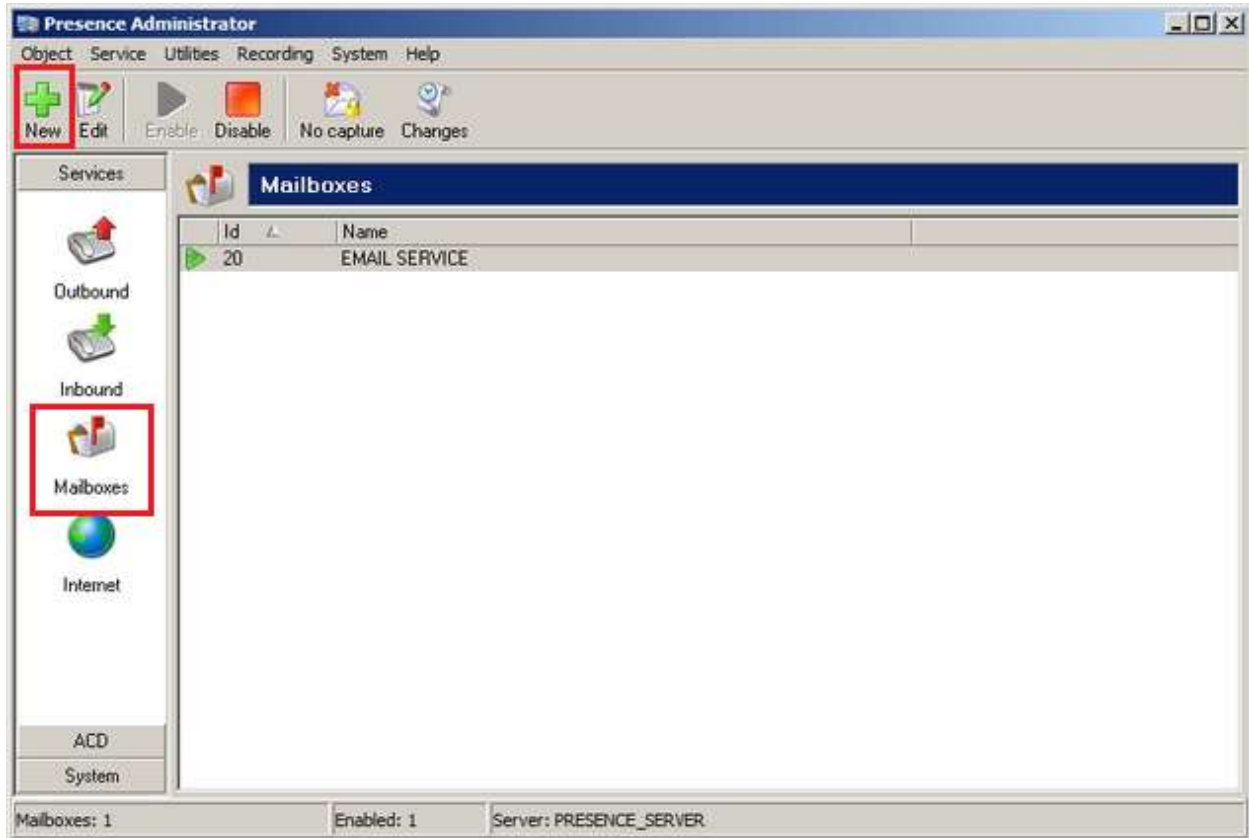
Select **Other** from the left hand side menu and moving to the right, select the **Enable direct transfer to agents of this service** check box. Enter the direct agent transfer VDN assigned in **Section 5.4.3** in the **Use the following VDN/CDN for transfer** field. Click **OK** to complete the inbound service configuration.

The screenshot shows the 'Inbound service' configuration window with the 'Other' tab selected. The left-hand menu lists various configuration options, with 'Other' highlighted. The main area contains three sections: 'After-call work', 'Transfer to agents', and 'Outgoing calls identification'. In the 'Transfer to agents' section, the checkbox 'Enable direct transfer to agents of this service' is checked and highlighted with a red rectangle. Below it, the dropdown menu 'Use the following VDN/CDN for transfer:' is set to '8274001'. The 'After-call work' section includes checkboxes for 'Minimum after-call work time' and 'Maximum after-call work time', each followed by a text input field and the unit 'seconds'. There is also a dropdown for 'Q: code for maximum time:' and a checkbox for 'Use q: code only if contact has not yet been qualified'. The 'Outgoing calls identification' section has a checkbox for 'Enable outgoing calls identification' and a text input field for 'Phone no:'. At the bottom right are 'OK' and 'Cancel' buttons.

Section	Option	Value / Status
After-call work	Minimum after-call work time	[ ] seconds
	Maximum after-call work time	[ ] seconds
	Q: code for maximum time	[ ]
	Use q: code only if contact has not yet been qualified	[ ]
Transfer to agents	Enable direct transfer to agents of this service	[x] (highlighted)
	Use the following VDN/CDN for transfer	8274001
Outgoing calls identification	Enable outgoing calls identification	[ ]
	Phone no:	[ ]

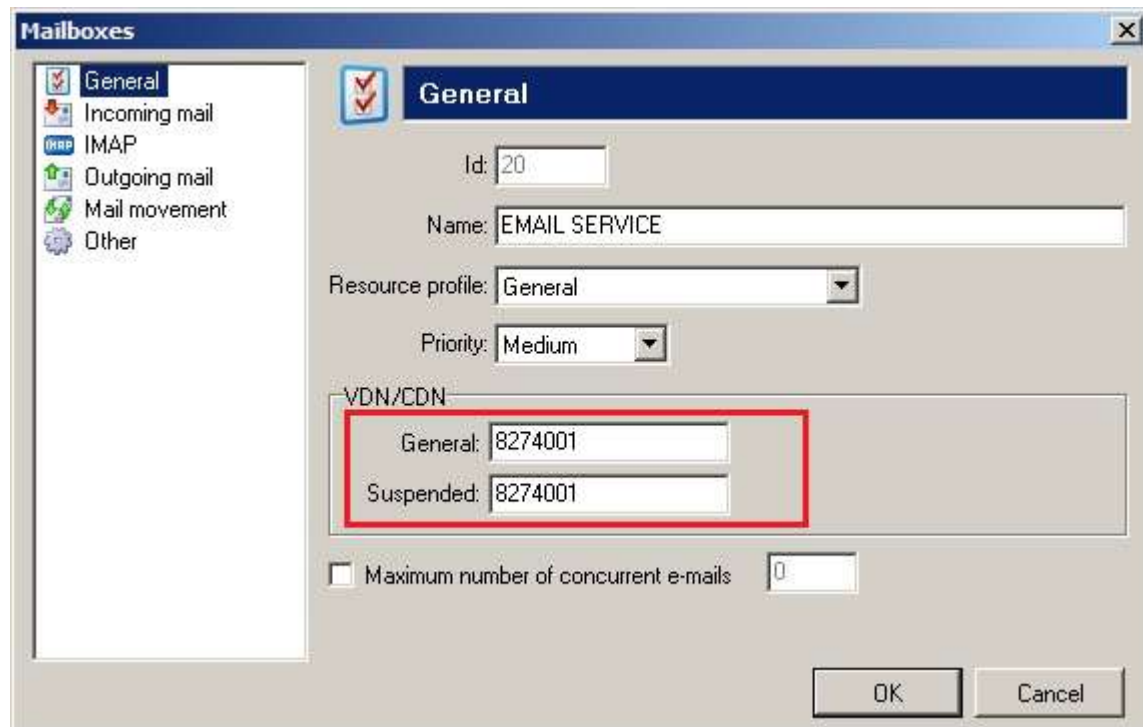
## 7.2.4 Email Service

To configure an email service, from the left hand side select **Services** → **Mailboxes** from the Presence Administrator main menu. Click the **New** button.





In the resulting screen, select **General** from the menu on the left hand side and enter a **Name** for the email service. Referring to **Section 5.5**, under the heading **VDN/CDN** in the **General** field enter the VDN assigned for email and enter the VDN assigned for suspended emails in the **Suspended** field. This is to allow each incoming email to be reported on. When the email arrives the VDN is called and the agent is placed on work.



The image shows a 'Mailboxes' configuration window with a sidebar on the left and a main configuration area on the right. The sidebar contains a list of categories: General (checked), Incoming mail, IMAP, Outgoing mail, Mail movement, and Other. The main area is titled 'General' and contains the following fields:

- Id:** 20
- Name:** EMAIL SERVICE
- Resource profile:** General (dropdown menu)
- Priority:** Medium (dropdown menu)
- VDN/CDN:**
  - General:** 8274001
  - Suspended:** 8274001
- ☐ Maximum number of concurrent e-mails: 0

At the bottom right are 'OK' and 'Cancel' buttons. A red rectangle highlights the 'General' and 'Suspended' VDN/CDN input fields.

Select **Incoming mail** from the left hand side menu. This window allows administrator to specify the POP3 server and account from which to download incoming mails. In the **Server** field enter the POP3 mail server address. For the interoperability testing this was the same IP address as the Presence Server. The POP3 port of **143** is entered into the **Port** field. Under the **Incoming mail account** heading enter the **Account name**, **Password** and **E-mail address** associated with the POP3 mail account.



The screenshot shows a Windows-style dialog box titled "Mailboxes" with a sub-tab titled "Incoming mail". On the left is a tree view with icons and labels: "General", "Incoming mail" (selected), "IMAP", "Outgoing mail", "Mail movement", and "Other". The main area is divided into two sections. The first section, "Incoming mail server (IMAP/POP3)", contains a "Protocol:" dropdown set to "IMAP", a "Server:" text box with "10.10.16.127", a "Port:" text box with "143", and a "Use the following type of encrypted connection:" dropdown set to "None" with a "Configure..." button. The second section, "Incoming mail account", contains three text boxes: "Account name:" with "server@prstestplans.com", "Password:" with "xxxxxxx", and "E-mail address:" with "server@prstestplans.com". At the bottom right are "OK" and "Cancel" buttons.

Select **Outgoing mail** from the left hand side menu and moving to the right, define the SMTP server that will be used to send response emails from Presence agents. Enter an IP address in the server field. For the interoperability testing this was the same IP address as the Presence Server. The SMTP port of **587** is entered into the **Port** field. Click **OK** to complete the email service configuration.

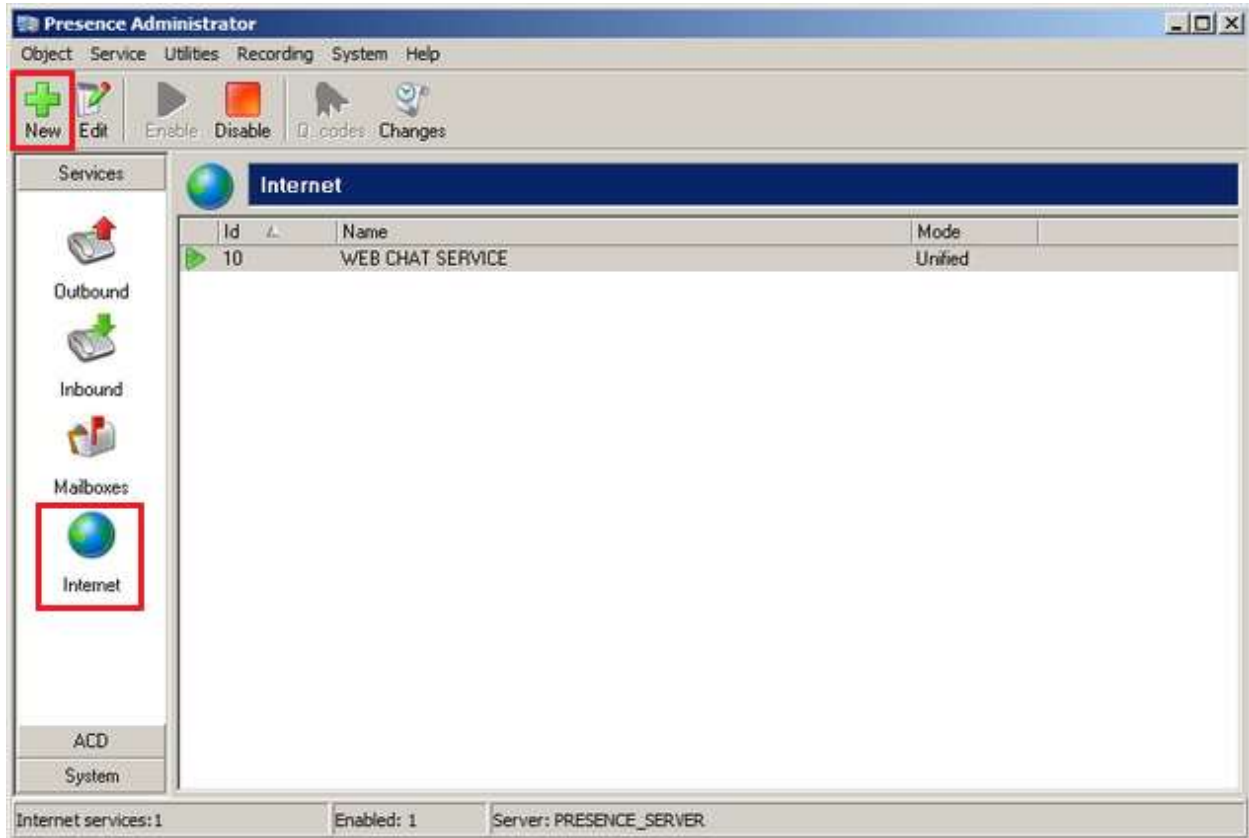
The screenshot shows the 'Mailboxes' configuration window with the 'Outgoing mail' tab selected. The left sidebar lists various mail settings: General, Incoming mail, IMAP, Outgoing mail (selected), Mail movement, and Other. The main area is titled 'Outgoing mail' and contains the following fields and options:

- Outgoing mail server (SMTP)**
  - Server:** 10.10.16.127
  - Port:** 587
  - Use the following type of encrypted connection:** None (dropdown menu)
  - Configure...** button
- ☒ **My server requires authentication**
  - ☒ Use same settings as my incoming mail server
  - ☐ Log on using
    - Account name:** [empty text box]
    - Password:** [empty text box]

At the bottom right, there are **OK** and **Cancel** buttons.

## 7.2.5 Web Chat / Web Call Back

To configure a web service, from the left hand side select **Services** → **Internet** from the Presence Administrator main menu. Click the **New** button.



In the resulting screen, select **General** from the menu on the left hand side and enter a **Name** for the web service. The **Enable chat service** and **Enable callback service** check boxes should be selected and the relevant VDN for each entered into the **VDN/CDN** field, click **OK** when done.

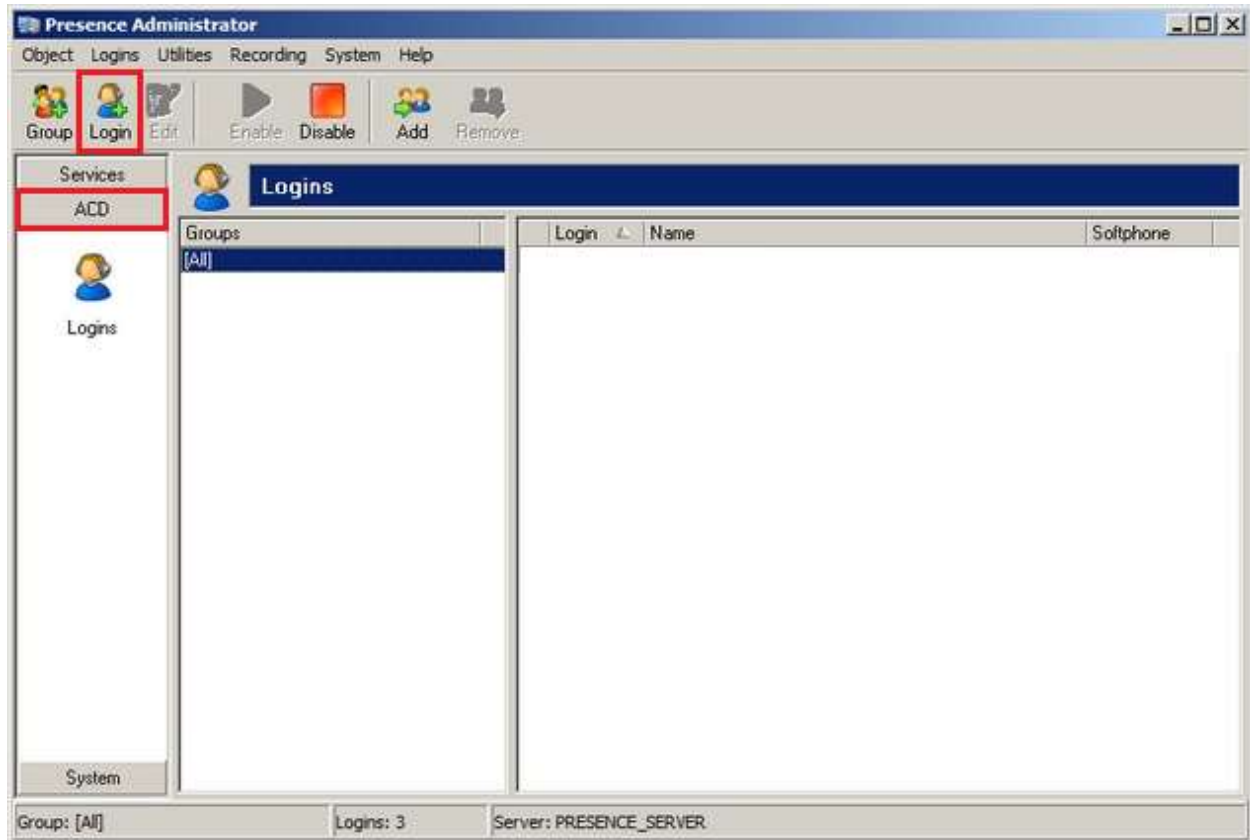
The screenshot shows the 'Internet service' configuration window with the 'General' tab selected. The left-hand menu lists various service categories: General, Web/URL, Logins, Interface, Texts, Mail, Service levels, and Other. The 'General' tab is active, showing the following fields and options:

- Id:** 10
- Name:** WEB CHAT SERVICE (highlighted with a red box)
- Mode:** Unified with other channels (dropdown menu)
- Chat section:**
  - ☒ Enable chat service (highlighted with a red box)
  - VDN:** 8274001 (highlighted with a red box)
- Callback section:**
  - ☒ Enable callback service (highlighted with a red box)
  - VDN:** 8274001 (highlighted with a red box)

At the bottom right, there are 'OK' and 'Cancel' buttons.

## 7.2.6 Add ACD Agent Logins

To add the agent logins administered on Communication Manager for use by Presence Suite, from the left hand pane of the Presence Administrator main menu select **ACD** → **Logins** and click the **Login** button.



In the **Logins** field, enter a Communication Manager Agent Login ID and a password, as configured in **Section 5.6**. Best practice is to tick **Agent cannot change password** as shown.

The screenshot shows the 'Insert logins' dialog box with the 'General' tab selected. The 'Logins' field contains the value '8271001'. The 'Password' field is masked with 'xxxxxxx', and the 'Confirm password' field is also masked with 'xxxxxxx'. The checkbox 'Agent cannot change password' is checked. The checkbox 'Use the agent password to log into the ACD system' is also checked. The 'Agent must change password at next login' checkbox is unchecked. The 'Password never expires' checkbox is checked. The 'OK' and 'Cancel' buttons are at the bottom right.

**Insert logins**

**General**

Logins: 8271001

Password: xxxxxxx

Confirm password: xxxxxxx

☐ Agent must change password at next login

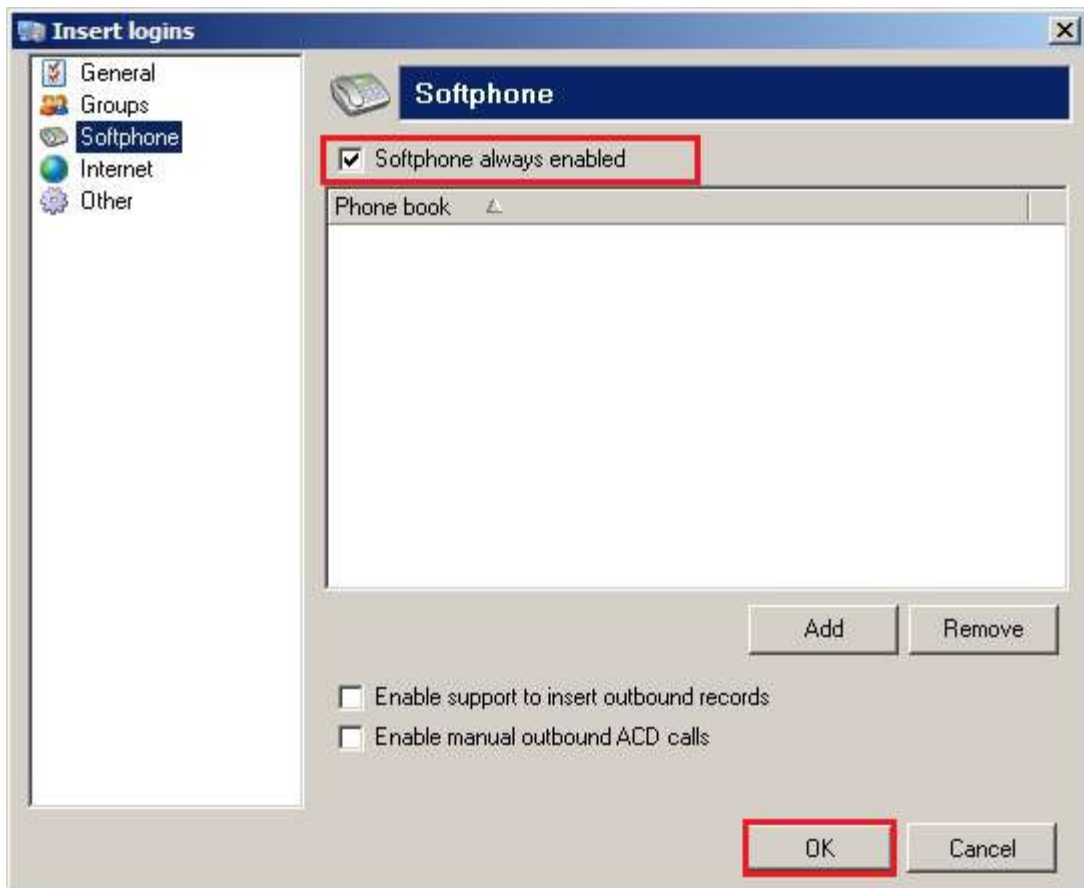
☒ Agent cannot change password

☒ Password never expires

☒ Use the agent password to log into the ACD system

OK Cancel

Click on **Softphone** in the left pane, and place a tick in the **Softphone always enabled** field. Click **OK** when done.





### 7.3 Presence Agent Configuration

The following steps are carried out on the Presence Suite Agent PC. Prior to installing the Presence agent, ensure that the DBExpress driver (dbexpoda.dll) is located in the **C:\Windows\System32** directory. The DBExpress driver allows the agent application to communicate with the Oracle database. Installing this driver eliminates the need to install the Oracle client. Launch the Presence agent configuration application by double clicking the **pcoagentcfg.exe** located in the **C: → Presence** folder. Enter the **Presence Server IP:** address as **10.10.16.127**. The **Presence Server port** can be left as the default value of **6100**. Enter the extension of the agent that will be using this workstation in the **Agent station** field. Check the **Hang up calls before logging in** check box. In the field **Use settings for** choose **Machine** from the drop down menu (not shown). Click **OK**. This step is needed for each agent configured; only the agent station field will vary.

The screenshot shows the 'Presence Agent Configuration' dialog box with the 'General' tab selected. The 'General' tab is highlighted in the left sidebar and the main content area. The 'Presence Server' section contains two text boxes: 'IP address' with the value '10.10.16.127' and 'Port' with the value '6100'. The 'Station configuration' section contains a text box for 'Agent station' with the value '8270001', a checked checkbox for 'Hang up calls before logging in', and an unchecked checkbox for 'Ask agent station at login window'. At the bottom, the 'Use settings for:' dropdown menu is set to 'Current user'. The 'OK' and 'Cancel' buttons are at the bottom right.

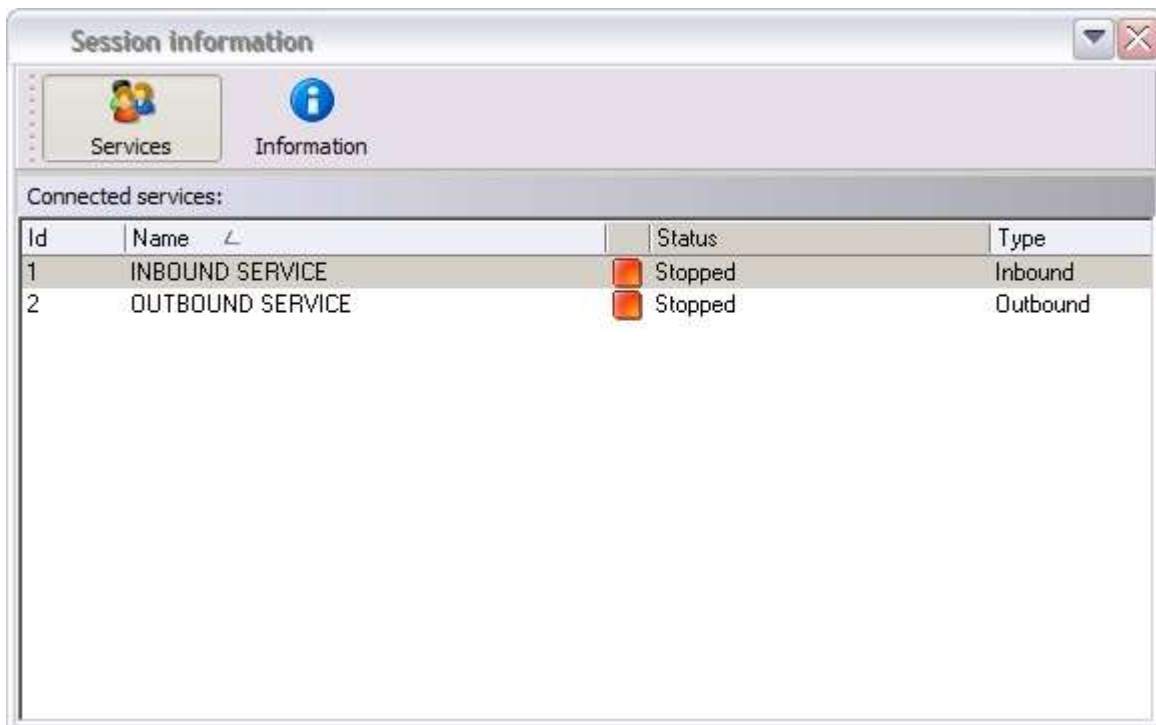
Field	Value
IP address	10.10.16.127
Port	6100
Agent station	8270001
Hang up calls before logging in	<input checked="" type="checkbox"/>
Ask agent station at login window	<input type="checkbox"/>
Use settings for	Current user

### 7.3.1 Logging in Presence Agent

Launch the Presence agent configuration application by double clicking the **pcoagent.exe** located in the Presence folder (not shown). Enter the agent **Login** and **Password** configured in **Section 5.6** and click on **OK**.



In the next screen, click on the **Services** button in the task bar. The service set up for the agent will be displayed.



A task bar is present at the top of the Agent PC. Click on the green arrow to put the agent in to an available state.



The information status on the task bar goes to available indicating the agent is ready to receive calls.



## 8. Verification Steps

This section provides the tests that can be performed to verify correct configuration of Communication Manager, Application Enablement Services and Presence Suite.

### 8.1 Verify Avaya Aura® Communication Manager CTI Link

The following steps can ensure that the communication between Communication Manager and the Application Enablement Services server is functioning correctly. Check the TSAPI link status with Application Enablement Services by using the command **status aesvcs cti-link**. Verify the **Service State** of the TSAPI link is **established**.

```
status aesvcs cti-link
```

AE SERVICES CTI LINK STATUS						
CTI Link	Version	Mnt Busy	AE Services Server	Service State	Msgs Sent	Msgs Rcvd
1	4	no	AES71678	established	87	61

Use the command **status aesvcs interface** to verify that the status **Local Node** of Application Enablement Services interface is connected and **listening**.

status aesvcs interface			
AE SERVICES INTERFACE STATUS			
Local Node	Enabled?	Number of Connections	Status
procr	yes	0	<b>listening</b>

Verify that there is a link with the Application Enablement Services and that messages are being sent and received by using the command **status aesvcs link**.

status aesvcs link						
AE SERVICES LINK STATUS						
Srvr/ Link	AE Services Server	Remote IP	Remote Port	Local Node	Msgs Sent	Msgs Rcvd
01/01	<b>AES71678</b>	<b>10.10.16.78</b>	<b>45883</b>	<b>procr</b>	<b>683</b>	<b>665</b>

## 8.2 Verify Avaya Aura® Application Enablement Services CTI Connection

The following steps are carried out on Application Enablement Services to ensure that the communication link between Communication Manager and the Application Enablement Services server is functioning correctly.

### 8.2.1 TSAPI Link

On the Application Enablement Services Management Console verify the status of the TSAPI link by selecting **Status → Status and Control → TSAPI Service Summary** to display the **TSAPI Link Details** screen. Verify the status of the TSAPI link by checking that the **Status** is **Talking** and the **State** is **Online**.

**TSAPI Link Details**

☐ Enable page refresh every 60 seconds

	Link	Switch Name	Switch CTI Link ID	Status	Since	State
	1	CM1627	1	Talking	Mon Nov 16 14:54:50 2015	Online

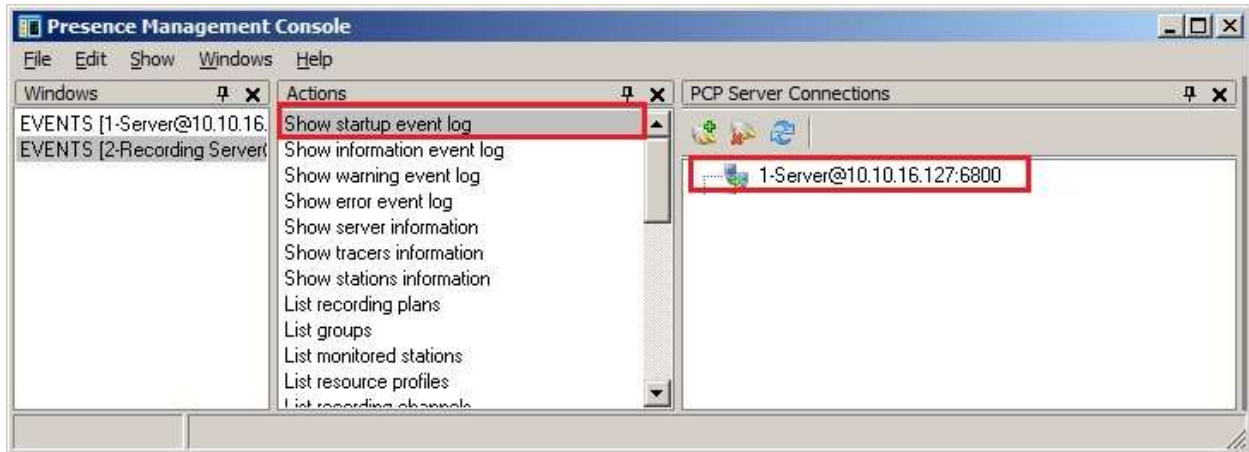
[Online](#) [Offline](#)

For service-wide information, choose one of the following:

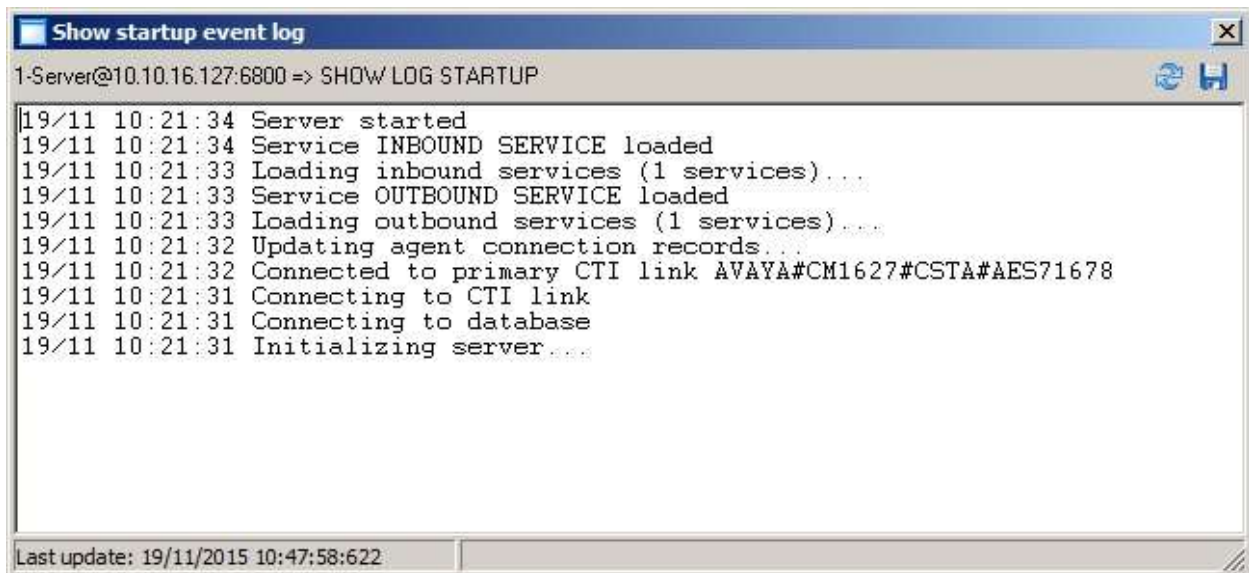
[TSAPI Service Status](#) [TLink Status](#) [User Status](#)

### 8.3 Verify Presence Suite CTI Connection

One of the available methods to confirm correct startup is a startup log which can be accessed from Presence Management Console. Navigate to **C: → Presence → pmconsole.exe** (not shown). A startup log commences when the Presence Server is trying to load and connect to the Application Enablement Services server. Click on the item named **Server@10.10.16.127:6800** in the **PCP Server Connections** pane of the Management Console. To open the startup event log, double click **Show startup event log** in the **Actions** pane.



Verify successful CTI connection and service startup.



## 9. Conclusion

These Application Notes describe the configuration steps required for Presence Suite R10.1 to successfully interoperate with Avaya Aura® Communication Manager R7.0 using Avaya Aura® Application Enablement Services R7.0. All feature functionality and serviceability test cases were completed successfully with observations noted in **Section 2.2**.

## 10. Additional References

This section references the Avaya and Presence Suite product documentation that are relevant to these Application Notes.

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

- [1] *Administering Avaya Aura® Communication Manager*, Document ID 03-300509
- [2] *Avaya Aura® Communication Manager Feature Description and Implementation*, Document ID 555-245-205
- [3] *Avaya Aura® Application Enablement Services Administration and Maintenance Guide Release 7.0*

The following documentation is available on request from Presence: [www.presenceco.com](http://www.presenceco.com)

- [4] *ACD Sys Presence Administrator Manual Presence Suite*, V10.1
- [5] *Presence Installation Guides Presence Software*, V10.1
- [6] *PBX/ACD Requirements Presence Software*, V10.1

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