

Avaya Solution & Interoperability Test Lab

Configuring SIP Connectivity between the Avaya Communication Server Integral 55 LX and Avaya Meeting Exchange Express Edition - Issue 1.0

Abstract

These Application Notes present the procedures for configuring SIP connectivity between the Avaya Communication Server Integral 55 LX (Avaya I55) and Avaya Meeting Exchange Express Edition (Avaya Meeting Exchange). This configuration leverages the flexibility offered by the Avaya I55 to support a rich set of audio conferencing options provided by Avaya Meeting Exchange.

1. Introduction

These Application Notes present the procedures for configuring SIP connectivity between the Avaya Communication Server Integral 55 LX (Avaya I55) and Avaya Meeting Exchange Express Edition (Avaya Meeting Exchange). This configuration leverages the flexibility offered by the Avaya I55 to support a rich set of audio conferencing options provided by Avaya Meeting Exchange.

Figure 1 illustrates the sample network configuration utilized for this compliance tested solution. The Avaya I55 is an ISDN based PABX (Private Automatic Branch Exchange) that also supports IP based clients and trunks via SIP and H.323. For this sample configuration, the Avaya I55 provided endpoint aggregation and media gateway functionality. For example, any station or trunk type associated with the Avaya I55 can interoperate with Avaya Meeting Exchange via SIP connectivity. The Avaya I55 also provided feature functionality for stations interoperating with Avaya Meeting Exchange, e.g., call hold, call transfer, three-way conference.

Avaya Meeting Exchange is a SIP-based voice conferencing solution that provides mid-market enterprise customers with an IP based audio conferencing system. For this sample configuration, Avaya Meeting Exchange was provisioned to accept calls from the Avaya I55 via either direct or basic call flows. A direct call flow allows access to conferences provisioned on Avaya Meeting Exchange without entering a passcode. Conversely, to enter a conference via a basic call flow requires a passcode. Avaya Meeting Exchange was also administered for outbound calling, which enabled call origination from Avaya Meeting Exchange to participants registered to the Avaya I55.

Signaling and media (audio) connectivity between the Avaya I55 and Avaya Meeting Exchange was configured for SIP and RTP respectively. Direct IP-to-IP audio connectivity on the Avaya I55 is not supported, e.g., the media exchange (RTP) between the Avaya 4600 Series IP Telephones registered to the Avaya I55 and Avaya Meeting Exchange is required to traverse the Avaya I55.

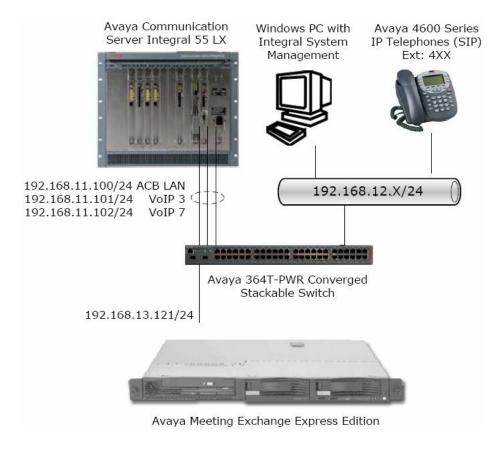


Figure 1: Sample Network Configuration

2. Equipment and Software Validated

The following equipment and software versions were used for this sample configuration:

Equipment	Software Version
Avaya Communication Server Integral 55 LX	
• ACB	L030V00.1.4
• VoIP Module(s)	VOIPSW56_rel_0166
Integral System Management	13.0.4
Avaya Meeting Exchange Express Edition	S6100-2.5.21.0
Avaya C364T-PWR Converged Stackable Switch	4.5.14
Avaya 4600 Series IP Telephones	2.2.2 (SIP)

Table 1: Equipment and Software Versions

3. Avaya Communication Server Integral 55 LX Configuration

This section displays the configuration for enabling the Avaya I55 to interoperate with Avaya Meeting Exchange.

The Avaya I55 is administered by means of Integral System Management (ISM), and its components. ISM is an application running on a PC with Windows-2000 or Windows-XP with IP connectivity to the Avaya I55. The following ISM components are used for the configuration:

- ICU Editor: A GUI based application for the administration of circuit packs on the system. In general, administration via the ICU Editor has the following template:
 - o Modify data.
 - o Save changes and exit provisioning screen.
 - o Send updates to the board.
 - o Reset board.
- Transparent Console (TCO) MML: A text based Command Line Interface (CLI) which can be used to administer the entire Avaya I55 system. However, for these Application Notes, TCO was used for administering areas that the ICU Editor could not. It should be noted that the procedures regarding MML commands are displayed in these Application Notes with liberal use of the "Help" command. To use Help, enter the question mark character "?", either at the command prompt, or appended to the end of a command. This will print a list of all commands available in the current mode. Using the Help command is optional; however, its use within these Application Notes better displays the options and syntax available to administrators.

3.1. Configure Integral System Management

This section describes the steps for configuring a customer account in ISM. A customer account in ISM is required to access and provision the Avaya I55.

Step **Description** Open ISM, and log in with the appropriate credentials. Administer settings for a customer 3.1.1 account as displayed. Enter a descriptive name for the trunk group in the **PABX** field. Select **IEE3** from the drop down menu for the **Software Version** field. Enter the appropriate credentials for the **Username**, password and password verification fields. Set the **Ethernet Interface** field to the IP address of the ACB LAN. Select the **Ethernet** radio button for the **TUX** field. Click the **Save** button to save the customer account. Click the **Select** button to open a connection to the Avaya I55. Customer Administration Customer: Sample Customer * New customer PABX: I55_AvayaMeetingExchange * New PABX Call Number Internal Software Version: IEE3 Address: -Access-C External + EDC @ Internal Username: C External C External + LDD Remote password: IP address Password verification: ISDN Interface MML Password Ethernet Interface 192 .168 .11 .100 Password verification: TUX PPP Password-

▼ Use Remote Password

Delete

PPP Password:

Password verification:

Ethernet

Select

C USB

C ISDN

C ISDN over Router

Cancel

Save

Test Connection

?

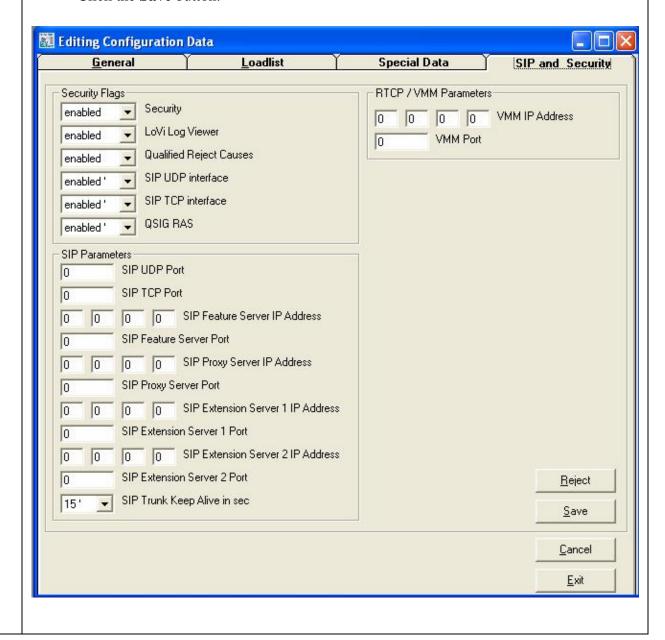
3.2. Configure VoIP Board for SIP Clients

This section describes the steps for configuring SIP signaling and media connectivity between the Avaya I55 and SIP clients. For this sample configuration, VoIP board 3 on the Avaya I55 was utilized for SIP clients, where SIP clients were Avaya 4600 Series IP Telephones. The configuration presented in this section is administered by means of the ICU Editor.

Step **Description** From ISM, open the ICU editor by selecting: **PABX Administration > SW Exchange** 3.2.1 Config Data → <Select VoIP Board> → Change Data → Execute. From the General tab, administer settings to enable codecs for SIP clients as displayed. Select 2 from the drop down menu for the **Number of Coder Groups** field. Select codecs from the drop down menu for the Coder Type field under Codergroup 1 and Codergroup 2 that are supported on the Avaya 4600 Series IP Telephones. Enter the appropriate IP network configuration for the board in the **VOIP Board IP** Address, Subnet Mask and Gateway IP Address fields. Use default settings for remaining fields. 💹 Editing Configuration Data Loadlist Special Data General ■ Number of Hybrid Channels Codergroup 2 G.729A ' Coder Type ■ Number of Coder Groups ■ Number of DSP Cores Codergroup 1 30ms:20kbps ' Packet Size G.711 ' ■ Number of DSP Cores FAX T38' Coder Type 20ms:80kbps ' Packet Size Number of DSP Cores Address from DHCP Server 101 VOIP Board IP Address 192 168 Reject Subnet Mask 255 255 255 0 Gateway IP Address 192 168 11 1 Save Cancel Exit

Step **Description** From the **Special Data** tab, administer settings for ethernet connectivity as displayed. 3.2.2 Administer appropriate ethernet configuration regarding physical connectivity under **Ethernet settings.** Use default settings for remaining fields. Editing Configuration Data General Loadlist SIP and Security **Special Data** IP configuration RTP packet loss message Overload control IP Port Range Threshold G.711 Max. registrations 1024' 🕶 2.5%' ▼ • IP Base Port Threshold G.729A Lower limit (%) 20000 6.0%' -Jitter Buffer Media Streaming Fax control FAX detection timer (sec) Decentralised M S Mode for min. size enabled ' Auto. 1 Minimum size in ms Keep Alive Timer Alternative Gatekeeper IP Phones in sec 10 10 10 3 IP Addr. Maximum size in ms 10' -300' Drop ratio G.711 QSIG Tunnel in sec 3.0%' ▼ Dynamic QISG RAS Port Dynamic QISG Drop ratio G.729A 3.0%' ▼ disabled ' Ethernet settings **RAS Port** Autonegotiation Type of Service ▼ Type of ToS definition Speed 100 Mbit normal ' ToS value Mode low delay fullduplex Error threshold Raw mode 2.0 %' 🔻 QSIG QoS Monitoring Telnet access QoS detection Telnet access disabled ' medium ' Threshold G.711 User name 2.5%' -Threshold G.729A 6.0%' ▼ Reject Threshold new calls (%) 70 Thresh, reactivate link 30 • <u>S</u>ave Cancel Exit

- **3.2.3** From the **SIP and Security** tab, administer settings to enable SIP connectivity for SIP clients as displayed.
 - Enable both the **SIP UDP interface** and **SIP TCP interface** fields from the drop down menu under **Security Flags**.
 - Enter **0** in both the **SIP UDP Port** and **SIP TCP Port** fields under **SIP Parameters**. *Note: The value 0 corresponds to port 5060.*
 - Use default settings for remaining fields.
 - Click the **Save** button.



3.3. Configure an Access Code for SIP Clients

This section describes the steps for configuring an access code for SIP clients on the Avaya I55. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step | **Description 3.3.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer settings to install an access code for SIP clients as follows: Enter 1:wabe; at the PROL< command prompt. To query available options, enter a "?" at the WABE< command prompt. ______ CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS ______ TC-03-3 26-04-2007 14:07:46 PROL (V4)192.168.12.144:1378 PROL<1:wabe; Command processing in progress ! WABE<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 pgwe - Program change */ /* 02 auge - Select output device */ /* 03 anzg - Branch to display menu */ $/\,$ 04 akze - Set up AKZ $^{\star}/\,$ /* 05 akzl - Clear AKZ */ /* 06 idke - Set up identification code */ /* 07 idkl - Clear identification code */ /* 08 cpnm - Install/delete company name */ /* 09 vbke - Set up VKZ */ /* 10 vbkl - Delete VKZ */ /* 11 datm - Branch to DAT DDI Alpha Tagging administration *//* 12 akza - Modify AKZ */

- **3.3.2** From TCO, configure AKZ to route a leading 4 plus 2 digits as follows:
 - To access the AKZ configuration menu, as well as query available options, enter **akze:?** at the **WABE**< command prompt.
 - To provision call routing for SIP clients, enter **4,intern,2,v**; at the **akze**:< command prompt. This command will route to **Dial group 2**, using the **Predial Dial method**, based on a **Selection code digit**, or leading digit "4" that is entered from a SIP client registered to the Avaya I55.
 - To query available options, enter a "?" at the **zids:** command prompt.
 - To complete the configuration of AKZ, enter ,2; at the **zids**:< command prompt. This setting defines the number of additional digits that are dialed to enable call routing.

```
WABE<akze:?
/* 01 akz - Selection code digit
              Format: 1..10 char.
              Poss. values: [0..F] */
   02 wsel - Dial selector
              Format: 2..6 charac.
              ANZG generates list of possible entries */
     03 wgrp - Dial group
              Poss. values: [1..128] / ALLE
              Block and specified entry possible
              Continuation permitted */
     04 wver - Dial method
              Poss. values: V = Predial / N = Postdial */
/* "E" means parameter input */
akze:<4,intern,2,v;
zids:<?
/* 01 bnnr - Bundle number of add. info data record
               Poss. values: [1..999]
              For selector GZL [A..F]
              Don't use bundle number 255! */
    02 akzi - AKZ information of add. info data record
              Poss. values: [-1..24] EXTERN selectors
              Poss. values: [-1..10] NETZ and QUE
               Poss. values: [0..4] KONTRA
              Poss. values: [0..8] other dialing sel. */
     03 lcds - LCR data
               optional parameter
               only in connection with bundle no.
              Default = 0 - no LCR data
              Poss. values: [0..4] */
    04 vwzz - Prefix allocation-Translation digit sequence
              No.of digits [1..10], Poss. values: [-1..9,A-D,*,#]
              Input without "," */
     05 vwzs - Prefix allocation-Translation selector
              Poss. values: [-1..1023]
              Meaning see Operating Instructions */
/* "E" means parameter input */
zids:<,2;
```

Step | Description **3.3.3** From TCO, verify the provisioning administered in this section as displayed. WABE<anzg; WABE<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 exit - Branch to first menu level */ /* 02 anzv - Display number of free reference blocks */ /* 03 dakz - Display data to an AKZ */ /* 04 dwse - Display data to a dial selector */ /* 05 dwgr - Display data to a dial group */ /* 06 dikz - Display data for identification code numbers */ /* 07 dise - Display data for identification selector */ /* 08 diwg - Display data for identification of a dial group */ /* 09 dcom - Display data of installed company no. */ /* 10 vbka - Display weighting data */ WABE<dwgr:2; 26.04.07 14:18:19 Display of dial evaluation data to a dial group Dial group : 2 Dial method : Predial Dial Bndl AKZ SA Co. LCR dialing ext. LCR RI- Num. sele. numb. Info group nr. data conversion all. rout SA Plan set digits sel cat. flg flg ______ EXTERN 3 0 INIT ROFF -Exch. line via earth NETZ 12 1 - - -0 INIT ROFF - -NETZ 12 1 -2 0 INIT ROFF -2 NETZ 12 1 3 NETZ 12 1 4 INTERN - 2 5 NETZ 12 1 6 NETZ 12 1 7 NETZ 12 1 81 NETZ 12 1 82 INTERN - 5 991 INTERN - 2 -0 INIT ROFF -0 - - -0 INIT ROFF -INIT ROFF -INIT ROFF -INIT ROFF -0 0 0 - - -0

3.4. Configure Lines for SIP Clients

This section describes the steps for configuring lines for SIP clients. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step | **Description 3.4.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer settings for SIP clients as follows: Enter 1:aogd; at the PROL< command prompt. To query available options, enter a "?" at the **AOGD**< command prompt. CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS ______ TC-03-3 26-04-2007 14:07:46 PROL (V4)192.168.12.144:1378 PROL<1:aogd; Command processing in progress ! AOGD<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 pgwe - Program change */ /* 02 anzg - Display AO basic data */ /* 03 aoae - Alter AO basic data */ /* 04 aoei - Set up AO basic data */ /* 05 aolo - Clear AO basic data */

3.4.2 From TCO, administer settings as follows:

- To access the AO configuration menu, as well as query available options, enter **aoei:?** at the **AOGD**< command prompt.
- To associate extension 401 with VoIP board 3, enter **401,01-01-03-16**; at the **aoei**:< command prompt.

```
AOGD<aoei:?
   01 aonr - Call No. or pseudo - Call No.
               Poss. values: 0..999999999 or
                              A0000..F9999 */
     02 hwad - AO-HWA
               format: GG-MD-ST-AO[-Z]
               group - modul - slot - AO of slot
[-Z]addition : 'S' second ICU adress ,'M' for MMG,
                               'C' for MSMC */
    03 bnnr - Bundle number for System Network AO
               Format: 1...3 digits
               Don't use bundle number 255! */
     04 task - Taskname for entry in RNPL
               Poss. entries:
               HOKO - Hotel communications (branch server)
                       (default call no. F9998)
               TKOM - Text communications
                       (default call no. F9997)
               IDAS - Information and data server
                       (default call no. F9996)
               ZETB - Central Electronic Directory
                       (default call no. F9995) */
/* "E" means parameter input */
aoei:<401,01-01-03-16;
Indicate AO type
```

3.4.3 From TCO, administer settings as displayed.

```
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 alae - Branch to additional alteration menu */
/* 05 aoty - Indicate or change AO type */
/* 06 dnei - Set up service */
/* 07 bkan - Indicate or alter B channel data */
AOGD<dnei:tlp;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 grda - Indicate or alter group data */
/* 05 ltgd - Indicate or alter line data */
/* 06 dnzu - Indicate or alter service data */
/* 07 dnei - Set up service */
/* 08 cdec - Coder / Echocanceler / Codermode install/change */
AOGD<grda:2,1;
AOGD<dnei:gen;
AOGD<dnzu:f;
AOGD<cdec:1,ein,n;
```

AOGD <anzg;< th=""><th></th><th></th><th></th><th></th><th></th></anzg;<>					
AUGD <anzg;< th=""><th></th><th></th><th></th><th></th><th></th></anzg;<>					
				26.04.07 14:3	32:12
Connecting circui	.t				
Call No.	: 40	=			
Slot / HWA		-01-03-16			
AO type	: DI	TN			
=======================================	:========	=========	========	=========	====
Service data					
	TLP	GEN			- 1
Status	RELEASED	+ RELEASED	+	+	+
	2	2	İ		i
	1	1	İ		i
Switchover group	0	0	İ		
Code dial group	0	0	İ		j
LCR-group	0	0			
	DEACTIVE	DEACTIVE			
Backward rel.		DEACTIVE			
	g711alaw64k		init	init	
	normal on	normal on	!	ļ	

3.4.5 From TCO, administer settings as follows:

- To return to the first menu, enter 1; at the **AOGD**< command prompt.
- To query available options, enter "?" at the **AOGD**< command prompt.
- Enter remaining commands as displayed.

```
AOGD<1;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 pgwe - Program change */
/* 02 anzg - Display AO basic data */
/* 03 aoae - Alter AO basic data */
/* 04 aoei - Set up AO basic data */
/* 05 aolo - Clear AO basic data */
AOGD<aoae;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 alae - Branch to additional alteration menu */
/* 05 aoty - Indicate or change AO type */
/* 06 dnei - Set up service */
/* 07 bkan - Indicate or alter B channel data */
/* 08 dnda - Alter service data */
/* 09 hwae - Alter AO HWA */
/* 10 nrae - Alter AO Call No. */
/* 11 rirr - Remote IP address / line call number install/change */
/* 12 lipi - Local IP Port / Connectivity Mode install/change */
AOGD<br/>bkan:all;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 bkan - Indicate or alter B channel data */
/* 07 bkzu - Indicate or alter B channel status */
AOGD<bkzu:f;
```

•	Description					
	From TCO, verify the provisioning adm	inistered thus far in this section as displayed.				
	Note: There is a maximum of two B-channels available for SIP clients.					
Ī	AOGD <anzg;< th=""></anzg;<>					
	26.04.07 14:25:18 Connecting circuit					
	Call No. : 401 Slot / HWA : 01-01-03-16 AO type : DITN					
B channel data						
Allocation code : - Deliberation code : - B chan. Bundle Direct Acc. Status B chan. Bundle Direct Acc. Status						
	B chan. Bundle Direct Acc. Status	B chan. Bundle Direct Acc. Status				
	number number	B chan. Bundle Direct Acc. Status				
		I I				
	number number	number number				
	number number 1 F Number of seizable B channels: 2 Seizure direction	number number 2 F				
	number number 1 F Number of seizable B channels: 2	number number 2 F Status B - BUSY D - DEFECT. EB - EDSS1 BUSY ER - EDSS1 RESERVED F - FREE				
	number number 1 F Number of seizable B channels: 2 Seizure direction	number number 2 F Status B - BUSY D - DEFECT. EB - EDSS1 BUSY ER - EDSS1 RESERVED				

3.4.7 From TCO, administer settings as displayed.

```
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 bkan - Indicate or alter B channel data */
/* 07 bkzu - Indicate or alter B channel status */
AOGD<aozu:f;
AOGD<aoae;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 alae - Branch to additional alteration menu */
/* 05 acty - Indicate or change AO type */
/* 06 dnei - Set up service */
/* 07 bkan - Indicate or alter B channel data */
/* 08 dnda - Alter service data */
/* 09 hwae - Alter AO HWA */
/* 10 nrae - Alter AO Call No. */
/* 11 rirr - Remote IP address / line call number install/change */
/* 12 lipi - Local IP Port / Connectivity Mode install/change */
AOGD<lipi:5060,11;
AOGD<alae;
```

Step | Description **3.4.8** From TCO, administer settings as displayed. AOGD<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 exit - Branch to first menu level */ /* 02 anzg - Display AO basic data */ /* 03 aozu - Indicate or change AO status */ /* 04 nako - Indicate or alter name / accounting section */ /* 05 vres - Indicate or alter reserved Connecting Memory */ /* 06 prve - Indicate or alter protocol / version */ /* 07 agrp - Indicate or alter general group data */ /* 08 uela - Indicate, alter, display overload priority */ /* 09 olae - Alter Operator overload entry */ AOGD<nako:SIP-ext1; AOGD<prve:etsi,0; AOGD<uela:2;

3.4.9 From TCO, verify the provisioning administered in this section as displayed.

```
AOGD<anzg;
                                                                  26.04.07 14:42:58
Connecting circuit
Call No.
                            : 401
Slot / HWA
AO type
                             : 01-01-03-16
                               : DITN
______
General ADS data
Accounting section : SIP-ext1
Protocols : SIP-ext1
                              Protocol | Version | faulty | busy 2 | error
                                  ETSI | 0 | OFF | OFF | OFF
Overload priority : 2
Public bar. unit gr. : 0
Colisee bar. unit gr. : 0
DISA-group : 0
Dealergroup : 0
CN alloc. HKZ line & tie :
Category : -1
Category : -1
Waiting field maximum : 0
Reserved
Connection memory : 0
Service memory : 2
AO state : IN OPERATION
Service block : sv-free
Call number block : Off
IP - address :
IP - address
-:-
Active Coder : - Secured registration : -
```

If additional SIP clients are required:

- Enter 1; at the AOGD< command prompt to return to the first menu.
- Repeat the steps in this section, changing only the extension number in the command entered in **Step 3.4.6**, e.g., AOGD<nako:SIP-ext1; to AOGD<nako:SIP-ext<n>;, where n is the next available number.

3.5. Configure Supported Services for SIP Clients

This section describes the steps for configuring supported services for SIP clients registered to the Avaya I55. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step Description

- **3.5.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer services for SIP clients as follows:
 - Enter 1:aolm; at the PROL< command prompt.
 - To query available options, enter a "?" at the **AOLM**< command prompt.
 - To administer services for the client associated with extension 401, enter **aoau:401**; at the **AOLM**< command prompt.
 - To query available options, enter a "?" at the **AOLM**< command prompt.
 - To enable AO related LM and variants for the client associated with extension 401, enter **falm:amt**; at the **AOLM**< command prompt.

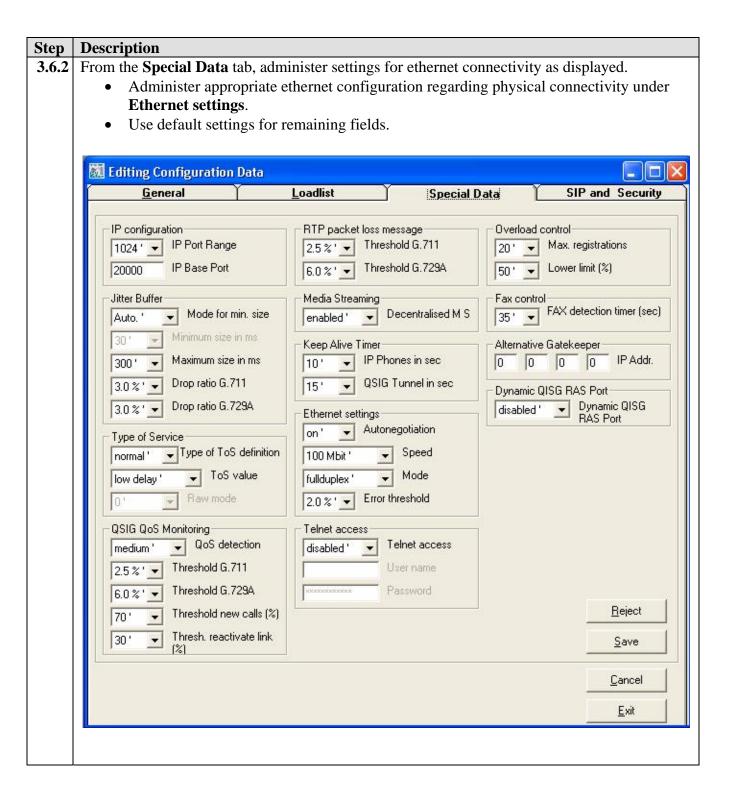
```
CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS
TC-03-3 23-05-2007 13:55:20 PROL
                                          (V4)192.168.12.144:2410
PROL<1:aolm;
Command processing in progress !
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 pgwe - Program change */
/* 02 auge - Select output device */
/* 03 aoau - Select AO number */
AOLM<aoau:401;
AOLM<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to the first program level */
/* 02 pgwe - Program change */
/* 03 dibe - Set service area */
/* 04 aalm - Display active LM and variants */
/* 05 falm - Enable AO related LM and variants */
/* 06 salm - Block AO related LM and variants */
AOLM<falm:amt;
```

Step	Description							
3.5.2	From TCO, verify the provisioning administered in this section as displayed.							
	AOLM <aalm;< th=""></aalm;<>							
	AO-Number	23.05.07 13:57:00 AO - Perform. features (Service: TLP)						
	400	AMT						
	AOLM<							
	If additional SIP clients require supported services:							
	• Enter 1; at the AOLM< command prompt to return to the first menu.							
	Repea	at the steps in this section for each client.						

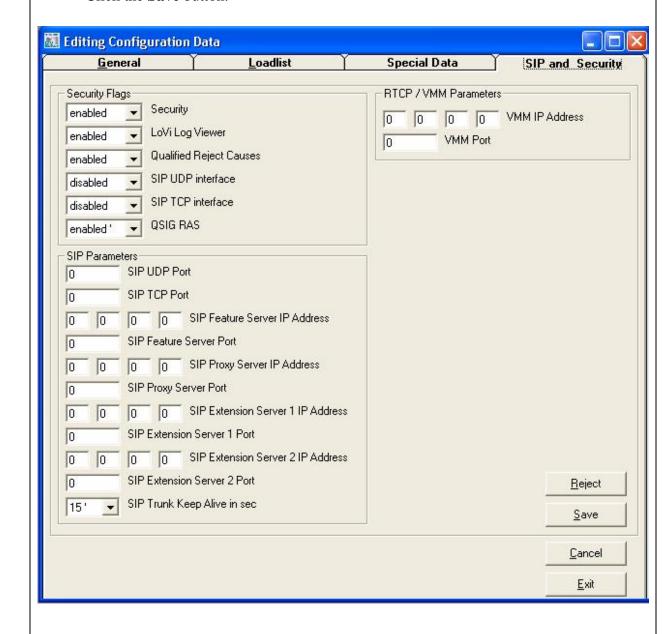
3.6. Configure VoIP Board for SIP Trunking

This section describes the steps for configuring SIP signaling and media connectivity between the Avaya I55 and Avaya Meeting Exchange. For this sample configuration, VoIP board 7 on the Avaya I55 was utilized for SIP trunking between the Avaya I55 and Avaya Meeting Exchange. The configuration presented in this section is administered by means of the ICU Editor.

Step **Description** From ISM, open the ICU editor by selecting: **PABX Administration > SW Exchange** 3.6.1 Config Data → <Select VoIP Board> → Change Data → Execute. From the General tab, administer settings to enable codecs for SIP trunking as displayed. Select 2 from the drop down menu for the **Number of Coder Groups** field. Select codecs from the drop down menu for the **Coder Type** field under **Codergroup 1** and Codergroup 2 that are supported on the Avaya 4600 Series IP Telephones. Enter the appropriate IP network configuration for the board in the **VOIP Board IP** Address, Subnet Mask and Gateway IP Address fields. Use default settings for remaining fields. Editing Configuration Data Loadlist Special Data SIP and Security General ■ Number of Hybrid Channels Codergroup 2-G.729A¹ ▼ Coder Type ■ Number of Coder Groups ■ Number of DSP Cores Codergroup 1 -30ms:20kbps¹ ▼ Packet Size G.711 Coder Type ▼ Number of DSP Cores FAX T38' - Coder Type 20ms:80kbps ' 🔻 Packet Size ■ Number of DSP Cores → Address from DHCP Server no 102 VOIP Board IP Address 11 192 168 Reject 255 255 0 Subnet Mask 255 192 168 11 1 Gateway IP Address Save Cancel Exit



- **3.6.3** From the **SIP and Security** tab, administer settings to enable SIP connectivity for SIP trunking as displayed.
 - Disable both the **SIP UDP interface** and **SIP TCP interface** fields from the drop down menu under **Security Flags**.
 - Use default settings for remaining fields.
 - Click the **Save** button.



3.7. Configure a Bundle

This section describes the steps for provisioning a bundle on the Avaya I55. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step | **Description 3.7.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer settings to provision a new bundle as follows: Enter 1:bndl; at the PROL< command prompt. To query available options, enter a "?" at the **BNDL**< command prompt. To display bundle data, enter anzg; at the BNDL< command prompt. CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS TC-03-3 26-04-2007 11:26:18 PROL (V4)192.168.12.144:1353 PROL<1:bndl; Command processing in progress ! /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 pgwe - Program change */ /* 02 auge - Select output device */ /* 03 anzg - Display bundle data */ /* 04 bnei - Set up bundle data */ /* 05 bnlo - Delete bundle data */ /* 06 bnae - Alter bundle data */ /* 07 vwza - Alter VWZ data to bundles */ /* 08 riae - Alter bundle seizure direction */ /* 09 skbt - Softkey and Bundle short text administration */ BNDL<anzg;

3.7.2 From TCO, administer settings as displayed.

```
BNDL<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 ands - Display number of free bundle data records */
/* 03 dueb - Display basic bundle of overflow bundle number */
/* 04 dbnd - Display data to bundles */
/* 05 dblt - Display Line Call Numbers to bundles */
/* 06 dbzu - Display bundle status */
/* 07 dbue - Display overflow bundle number to bundles */
/* 08 dvwz - Display VWZ data to bundles */
/* 09 aran - Display FOAC reactions for bundles */
BNDL<dblt:<?
/* 01 bnnr - Bundle number
               Poss. values: [1..999] / ALLE
               Don't use bundle number 255!
               Block and specified entry possible
              Continuation permitted */
/* Digit number - [01,20] */
BNDL<dblt:<alle;</pre>
                                                             26.04.07 11:29:21
Bundle number
Lines entered :
Bundle number
Lines entered:
Bundle number
Lines entered :
Bundle number
Lines entered :
Bundle number
                      : 13
Lines entered :
                 : 200
Bundle number
Lines entered :
BNDL<exit;
```

3.7.3 From TCO, administer settings as displayed.

```
BNDL<bnei:?
/* 01 bnnr - Bundle number
               Poss. values: [1..999] / ALLE
               Don't use bundle number 255!
               Block and specified entry possible
               Continuation permitted */
   02 uenr - Overflow bundle number
               Poss. values:[0..999]
               Don't use bundle number 255!
               0 = No overflow bundle */
/* "E" means parameter input */
bnei:<86,;
BNDL<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 bnzu - Alter bundle status */
/* 03 uebn - Alter overflow bundle number */
/* 04 alva - Alter bundle ALV flag */
/* 05 stke - Alter bundle ALV stage */
/* 06 ltza - Alter bundle line limit */
/* 07 lwzf - Alter route digit sequence */
/* 08 nwfl - Change postdialing flag */
/* 09 bntx - Information text change */
/* 10 wbna - Alter other bundle data */
/* 11 qlfl - Change QSIG-Line flag */
BNDL<bnzu:?
/* 01 zust - Status
              Poss. values: s/f */
/* "E" means parameter input */
bnzu:<f;
BNDL<bnae:<86;
BNDL<stke:?
/* 01 keng - ALV stage
               Format: 4 charac.
               ANZG generates list of possible entries */
/* "E" means parameter input */
stke:<anzg;
Brief staging designator :
INTR -
          Intern
AMTK -
         Exchange Line incomming
AMTG -
         Exchange Line outgoing
LMUE -
          LM overlapping
stke:<amtk;
```

Step | Description **3.7.4** From TCO, verify the provisioning administered in this section as displayed. BNDL<anzg; BNDL<dbnd:<86; 26.04.07 11:48:54 : 86 Bundle number Overflow bundle number : 0 Bundle status : free Bundle type : -- ALV stage : ext.incoming Seizure direction : init init : 0 : 0 Line limit Route digit sequence : -Route digit sequence Postdialing flag : nein : no Information text : - VWZ data outgoing: Allocation digit : -1,-1,-1,-1,-1,-1,-1,-1,-1 Selector : -1 incoming: Allocation digit : -1,-1,-1,-1,-1,-1,-1,-1,-1 Selector : -1 FOAC reactions for bundle number Event Reaction nstf nein nstb nein kres nein nein uvwa newa nein nbao nein nein anaw nein kbne nein nein aoab nuel Lines entered : Explanation for FOAC data - FOAC event: NSTF - called extension free KRES - no resources available NEWA - invalid dialling ANAW - call rejected AOAB - out of order NSTB - called extension busy UVWA - incomplete dialling NBAO - called extension not assigned KBNE - incoming calls barred NUEL - network busy AOAB - out of order NUEL - network busy Explanation for FOAC data - FOAC reaction: APSO - forward to AC immediately APNZ - forward to AC after timeout AUSL - release call NEIN - no FOAC specific reaction (treatment as if FOAC was not active)

3.8. Configure an Access Code for SIP Trunking

This section describes the steps for configuring an access code for SIP trunking on the Avaya I55. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step | **Description 3.8.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer settings to install an access code for SIP clients as follows: Enter 1:wabe; at the PROL< command prompt. To query available options, enter a "?" at the WABE< command prompt. CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS ______ TC-03-3 26-04-2007 11:26:18 PROL (V4)192.168.12.144:1353 PROL<1:wabe; Command processing in progress ! WABE<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 pgwe - Program change */ /* 02 auge - Select output device */ /* 03 anzg - Branch to display menu */ $/\,$ 04 akze - Set up AKZ $^{\star}/\,$ /* 05 akzl - Clear AKZ */ /* 06 idke - Set up identification code */ /* 07 idkl - Clear identification code */ /* 08 cpnm - Install/delete company name */ /* 09 vbke - Set up VKZ */ /* 10 vbkl - Delete VKZ */ /* 11 datm - Branch to DAT DDI Alpha Tagging administration *//* 12 akza - Modify AKZ */

3.8.2 From TCO, administer settings as displayed.

```
WABE<akze:?
/* 01 akz
            - Selection code digit
               Format: 1..5 charac. for dial selector KONTRA
               Format: 1..10 charac. for dial selector QUE, KNR, NETZ,
                                     OUTCC, INTAKZ
               Format: 1..9 charac. for all other dial selectors
               Poss. values: [0..F] */
     02 wsel - Dial selector
               Format: 2..6 charac.
               ANZG generates list of possible entries */
     03 wgrp - Dial group
               Poss. values: [1..128] / ALLE
               Block and specified entry possible
               Continuation permitted */
     04 wver - Dial method
               Poss. values: V = Predial / N = Postdial */
/* "E" means parameter input */
akze:<e0,extern,2,v;
zids:<?
   01 bnnr - Bundle number of add. info data record
               Poss. values: [1..999]
               For selector GZL [A..F]
               Don't use bundle number 255! */
     02 akzi - AKZ information of add. info data record
               Poss. values: [-1..24] EXTERN selectors
               Poss. values: [-1..10] NETZ and QUE
               Poss. values: [0..4] KONTRA
               Poss. values: [0..8] other dialing sel. */
     03 lcds - LCR data
               optional parameter
               only in connection with bundle no.
               Default = 0 - no LCR data
               Poss. values: [0..4] */
     04 vwzz - Prefix allocation-Translation digit sequence
               No.of digits [1..10], Poss. values: [-1..9,A-D,*,#]
               Input without "," */
     05 vwzs - Prefix allocation-Translation selector
               Poss. values: [-1..1023]
               Meaning see Operating Instructions */
/* "E" means parameter input */
zids:<86,3;
zids:<;
```

3.8.3 From TCO, administer settings as displayed.

```
WABE<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 pgwe - Program change */
/* 02 auge - Select output device */
/* 03 anzg - Branch to display menu */
/* 04 akze - Set up AKZ */
/* 05 akzl - Clear AKZ */
/* 06 idke - Set up identification code */
/* 07 idkl - Clear identification code */
/* 08 cpnm - Install/delete company name */
/* 09 vbke - Set up VKZ */
/* 10 vbkl - Delete VKZ */
/* 11 datm - Branch to DAT DDI Alpha Tagging administration */
/* 12 akza - Modify AKZ */
WABE<akzl:?
    01 akz - Selection code digit
               Format: 1..10 char.
               Poss. values: [0..F] */
     02 wsel - Dial selector
               Format: 2..6 charac.
               ANZG generates list of possible entries */
     03 wgrp - Dial group
               Poss. values: [1..128] / ALLE
               Block and specified entry possible
               Continuation permitted */
     04 wver - Dial method
               Poss. values: V = Predial / N = Postdial */
/* "E" means parameter input */
akz1:<e0;
WABE<akze:?
/* 01 akz - Selection code digit
               Format: 1..5 charac. for dial selector KONTRA
               Format: 1..10 charac. for dial selector QUE, KNR, NETZ,
                                       OUTCC, INTAKZ
               Format: 1..9 charac. for all other dial selectors
               Poss. values: [0..F] */
     02 wsel - Dial selector
               Format: 2..6 charac.
               ANZG generates list of possible entries */
     03 wgrp - Dial group
               Poss. values: [1..128] / ALLE
               Block and specified entry possible
               Continuation permitted */
     04 wver - Dial method
               Poss. values: V = Predial / N = Postdial */
/* "E" means parameter input */
akze:<e0,netz,alle,v;</pre>
```

3.8.4 From TCO, administer settings as displayed.

```
zids:<?
    01 bnnr - Bundle number of add. info data record
              Poss. values: [1..999]
              For selector GZL [A..F]
               Don't use bundle number 255! */
    02 akzi - AKZ information of add. info data record
               Poss. values: [-1..24] EXTERN selectors
               Poss. values: [-1..10] NETZ and QUE
               Poss. values: [0..4] KONTRA
              Poss. values: [0..8] other dialing sel. */
    03 lcds - LCR data
               optional parameter
               only in connection with bundle no.
              Default = 0 - no LCR data
              Poss. values: [0..4] */
    04 vwzz - Prefix allocation-Translation digit sequence
              No.of digits [1..10], Poss. values: [-1..9,A-D,*,#]
              Input without "," */
    05 vwzs - Prefix allocation-Translation selector
              Poss. values: [-1..1023]
              Meaning see Operating Instructions */
/* "E" means parameter input */
zids:<86,3;
zids:<;
```

Step | **Description 3.8.5** From TCO, verify the provisioning administered in this section as displayed. WABE<anzg; WABE<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 exit - Branch to first menu level */ /* 02 anzv - Display number of free reference blocks */ /* 03 dakz - Display data to an AKZ */ /* 04 dwse - Display data to a dial selector */ /* 05 dwgr - Display data to a dial group */ /* 06 dikz - Display data for identification code numbers */ /* 07 dise - Display data for identification selector */ /* 08 diwg - Display data for identification of a dial group */ /* 09 dcom - Display data of installed company no. */ /* 10 vbka - Display weighting data */ WABE<dwgr:2,v; 04.05.07 15:50:55 Display of dial evaluation data to a dial group _____ Dial group : 2 Dial method : Predial Dial Bndl AKZ SA Co. LCR dialing ext. LCR RI- Num. sele. numb. Info group nr. data conversion all. rout SA Plan set digits sel cat. flg flg ______ EXTERN 3 0 INIT ROFF -Exch. line via earth NETZ 12 1 - - -NETZ 12 1 - - -0 INIT ROFF - -2 0 INIT ROFF -NETZ 12 1 NETZ 12 1 INTERN - 2 NETZ 12 1 NETZ 12 1 NETZ 12 1 NETZ 12 1 NETZ 12 1 INTERN - 5 INTERN - 2 RUVA - -3 0 INIT ROFF -0 4 - - -0 INIT ROFF -1 5 INIT ROFF -INIT ROFF -6 0 0 7 INIT ROFF -0 81 82 0 - - -0 991 Α PUALLG -C CW -D EO NETZ 86 3 - - - 0 INIT ROFF - -

3.9. Configure Lines for SIP Trunking

This section describes the steps for configuring lines for SIP trunking. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step | **Description 3.9.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer settings for SIP clients as follows: Enter 1:aogd; at the PROL< command prompt. To query available options, enter a "?" at the **AOGD**< command prompt. CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS ______ TC-03-3 26-04-2007 11:26:18 PROL (V4)192.168.12.144:1353 PROL<1:aogd; Command processing in progress ! AOGD<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 pgwe - Program change */ /* 02 anzg - Display AO basic data */ /* 03 aoae - Alter AO basic data */ /* 04 aoei - Set up AO basic data */ /* 05 aolo - Clear AO basic data */

3.9.2 From TCO, administer settings as follows:

- To access the AO configuration menu, as well as query available options, enter **aoei:?** at the **AOGD**< command prompt.
- To associate SIP trunking with VoIP board 7, enter **c9000,01-01-07-30**; at the **aoei**:< command prompt.

```
AOGD<aoei:?
   01 aonr - Call No. or pseudo - Call No.
               Poss. values: 0..999999999 or
                              A0000..F9999 */
     02 hwad - AO-HWA
               format: GG-MD-ST-AO[-Z]
               group - modul - slot - AO of slot
[-Z]addition : 'S' second ICU adress ,'M' for MMG,
                               'C' for MSMC */
    03 bnnr - Bundle number for System Network AO
               Format: 1...3 digits
               Don't use bundle number 255! */
     04 task - Taskname for entry in RNPL
               Poss. entries:
               HOKO - Hotel communications (branch server)
                       (default call no. F9998)
               TKOM - Text communications
                       (default call no. F9997)
               IDAS - Information and data server
                       (default call no. F9996)
               ZETB - Central Electronic Directory
                       (default call no. F9995) */
/* "E" means parameter input */
aoei:<c9000,01-01-07-30;
Indicate AO type
```

3.9.3 From TCO, administer settings to associate 30 B-channels with VoIP board 7 as displayed.

```
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 alae - Branch to additional alteration menu */
/* 05 acty - Indicate or change AO type */
/* 06 dnei - Set up service */
/* 07 bkan - Indicate or alter B channel data */
AOGD<aoty:?
   01 typ - AO Type
               Format: 3..4 char.
               ANZG generates list of possible entries */
     02 plty - Special position functions for AO type DIPL
                Poss. values: [ J / N ]
                               = J (dealer or ACD)
                Spec.pos.:
                Operator pos.: = N */
/* "E" means parameter input */
aoty:<pra;
AO blocked and out of order
AOGD<bkan:?
   01 bknr - B channel no.
                Indic. of B channel no.
                Poss. values: [1..30] / ALL
                Block and specified entry possible */
    02 vgke - Allocation code
                Format: 4 char.
                Poss. entries: NSTA - PABX
                                GEGA - Opp. system */
     03 vhke - B channel deliberation code
                Poss. values: J - B channel deliberation possible
                               N - B channel deliberation not poss. */
/* "E" means parameter input */
bkan:<all,nsta;
```

3.9.4 From TCO, administer settings as displayed.

```
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 bkan - Indicate or alter B channel data */
/* 05 ribn - Indicate or alter seizure direction and bundle allocation */
/* 06 zugr - Indicate or alter primary access right */
/* 07 bkzu - Indicate or alter B channel status */
AOGD<ribn:?
/* 01 rich - Seizure direction
                Poss. value : G - outgoing
                              K - incoming
                              W - bothway */
    02 bnnr - Bundle number
             - Bundle for analogue and digital trunk lines [1..999]
                  Don't use bundle 255!
              - Bundle for announcement [1..10]
                      1 = ACD, 2 = HOKO, 3 = common,
                      4 = ACD calls on hold, 5..9 = unused,
                      10 = recording ACD announcem.
                Format: 1...3 digits */
/* "E" means parameter input */
ribn:<w,86;
AOGD<bkzu:f;
```

Step | Description **3.9.5** From TCO, verify the provisioning administered thus far in this section as displayed. *Note*: There is a maximum of thirty B-channels available for SIP trunking. AOGD<anzg; ______ 26.04.07 12:43:26 Connecting circuit Call No. Slot / HWA AO type : C9000 - C9029 : 01-01-07-30 : PRA ______ B channel data : NSTA Allocation code Deliberation code : DEACTIVE B chan. Bundle Direct Acc. Status | B chan. Bundle Direct Acc. Status number number number number 1 86 W O F 16 86 W O F 2 86 W O F 17 86 W O F 3 86 W O F 18 86 W O F 4 86 W O F 19 86 W O F 5 86 W O F 20 86 W O F 6 86 W O F 21 86 W O F 7 86 W O F 22 86 W O F 8 86 W O F 22 86 W O F 8 86 W O F 22 86 W O F 9 86 W O F 23 86 W O F 9 86 W O F 24 86 W O F 10 86 W O F 24 86 W O F 11 86 W O F 25 86 W O F 11 86 W O F 26 86 W O F 12 86 W O F 26 86 W O F 13 86 W O F 27 86 W O F 14 86 W O F 28 86 W O F 15 86 W O F 29 86 W O F Number of seizable B channels: 30 Seizure direction Status G - outgoing B - BUSY K - incoming D - DEFECT. W - bothway EB - EDSS1 BUSY ER - EDSS1 RESERVED F - FREE G - FAULTY R - RESERVED Access right ----- | S - BARRED M - with T - DEFECT./BARRED O - without V - SEIZED/BARRED ______

3.9.6 From TCO, administer settings as displayed.

```
AOGD<1;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 pgwe - Program change */
/* 02 anzg - Display AO basic data */
/* 03 aoae - Alter AO basic data */
/* 04 aoei - Set up AO basic data */
/* 05 aolo - Clear AO basic data */
AOGD<aoae;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 alae - Branch to additional alteration menu */
/* 07 bkan - Indicate or alter B channel data */
/* 08 dnda - Alter service data */
/* 09 hwae - Alter AO HWA */
/* 10 nrae - Alter AO Call No. */
/* 11 rirr - Remote IP address / line call number install/change */
/* 12 lipi - Local IP Port / Connectivity Mode install/change */
AOGD<dnda:?
/* 01 dnst - Service
               Format : 3 char.
               Default: TLP - Telephony
               ANZG generates list of possible entries */
/* "E" means parameter input */
dnda:<tlp;</pre>
```

3.9.7 From TCO, administer settings as displayed.

```
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 grda - Indicate or alter group data */ /* 05 ltgd - Indicate or alter line data */
/* 06 dnzu - Indicate or alter service data */
/* 07 dnei - Set up service */
/* 08 dnda - Alter service data */
/* 09 dnlo - Clear service */
/* 10 cdec - Coder / Echocanceler / Codermode install/change */
AOGD<grda:?
    01 wgrp - Dial group
               Format: 1...3 digits */
     02 vgrp - Traffic group
               Format: 1..2 digits */
    03 ugrp - Switchover group
               Format: 1 digit */
     04 cwgr - Code dialing group
               Format: 1..3 digits */
     05 lcrg - Least Cost Routing (LCR)-group
               Format: 1..2 digits */
/* "E" means parameter input */
grda:<2,1,0,0,0;
AOGD<cdec:?
/* 01 code - Codername
               Input format :
               nnn - nnn 1 to 24 characters
               Possible values with ANZG */
     02 echo - Echocanceler
               Possible values :
               EIN / ON
               AUS / OFF */
     03 como - Codermode
               Possible values :
               N = normal
               T = transparent */
/* "E" means parameter input */
cdec:<1,ein,n;
AOGD<dnzu:f;
```

AOGD <anzg;< th=""></anzg;<>										
26.04.07 12:48:52 Connecting circuit										
Call No. : C9000 - C9029 Slot / HWA : 01-01-07-30 AO type : PRA										
Service data										
ļ	TLP		ļ.	ļ.	1					
Dial group Traffic group Switchover group Code dial group LCR-group Dial retrieval	RELEASED 2 1 0 0 0 DEACTIVE DEACTIVE g711alaw64k normal on	 init	init	init						

Step | **Description 3.9.9** From TCO, administer settings as displayed. AOGD<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 exit - Branch to first menu level */ /* 02 anzg - Display AO basic data */ /* 03 aozu - Indicate or change AO status */ /* 04 grda - Indicate or alter group data */ /* 05 ltgd - Indicate or alter line data */ /* 06 dnzu - Indicate or alter service data */ /* 07 dnei - Set up service */ /* 08 dnda - Alter service data */ /* 09 dnlo - Clear service */ /* 10 cdec - Coder / Echocanceler / Codermode install/change */ AOGD<dnei:gen; AOGD<dnda:gen; AOGD<cdec:<11,ein,n; AOGD<grda:7,0,0,0,0; WARNING - No traffic group entered for service 19 AOGD<dnzu:f;

Step Description

3.9.10 From TCO, administer settings as displayed.

AOGD<1;
AOGD<aoae;
AOGD<?
/* SE Status display switch on */

```
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 04 alae - Branch to additional alteration menu */
/* 05 acty - Indicate or change AO type */
/* 06 dnei - Set up service */
/* 07 bkan - Indicate or alter B channel data */
/* 08 dnda - Alter service data */
/* 09 hwae - Alter AO HWA */
/* 10 nrae - Alter AO Call No. */
/* 11 rirr - Remote IP address / line call number install/change */
/* 12 lipi - Local IP Port / Connectivity Mode install/change */
AOGD<alae;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzq - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 nako - Indicate or alter name / accounting section */
/* 05 vres - Indicate or alter reserved Connecting Memory */
/* 06 prve - Indicate or alter protocol / version */
/* 07 agrp - Indicate or alter general group data */
/* 08 uela - Indicate, alter, display overload priority */
/* 09 olae - Alter Operator overload entry */
AOGD<nako:SIP_Trunk;
```

3.9.11 From TCO, administer settings as displayed.

```
AOGD<vres:?
   01 rese - Number of Connecting Memories
              Poss. values: [0..2] */
     02 wfdm - waiting field maximum
              possible values: 0..10 */
/* "E" means parameter input */
vres:<,10;
AOGD<uela:?
/* 01 prio - Priority
               Poss. values: [0..2] */
     02 pran - Display of priority distribution to AO group
               Poss. value: a */
/* "E" means parameter input */
uela:<2;
AOGD<agrp:?
   01 spam - Public Barring Unit group
              Format: 1...2 digits */
     02 grkn - Enter group allocation for HKZ line
               For dealer group: [G0..G255]
                                  G0 = deleted
           or
               Call number allocation for HKZ line and tie
               Poss. values: 0..999999999 or A0000..F9999
                             L = delete */
     03 digr - DISA-group
               Format: 1 digit */
     04 katg - category
               possible values: -1..9 */
     05 spco - Barring Unit COLISEE NUMERIS
              Format: 1..2 digit */
/* "E" means parameter input */
agrp:<1,,,-1;
AOGD<prve:?
   01 prot - Protocol
               Poss. values: INIT, 8030, 1TR6, TN1R6, ECMA, ETSI, NTA, DKZN1,
                             DKZN2, QSIG, STIM, MOBI, NI */
     02 vers - Version
               Poss. values: [0..255]
               Enter both, Protocol and Version
               Delete if Protocol and Version are present */
     03 lmff - Release LM
               Format: 4 characters
               ANZG generates list of possible entries */
     04 lmfs - Block LM
               Format: 4 characters
               ANZG generates list of possible entries */
/* "E" means parameter input */
prve:<etsi,60;</pre>
```

3.9.12 From TCO, administer settings as displayed.

```
AOGD<aoae;
AOGD<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Branch to first menu level */
/* 02 anzg - Display AO basic data */
/* 03 aozu - Indicate or change AO status */
/* 04 alae - Branch to additional alteration menu */
/* 05 aoty - Indicate or change AO type */
/* 06 dnei - Set up service */
/* 07 bkan - Indicate or alter B channel data */
/* 08 dnda - Alter service data */
/* 09 hwae - Alter AO HWA */
/* 10 nrae - Alter AO Call No. */
/* 11 rirr - Remote IP address / line call number install/change */
/* 12 lipi - Local IP Port / Connectivity Mode install/change */
AOGD<rirr:?
   01 ipvr - Remote IP version
               Posible values : V4 or V6 */
     02 ipad - Remote IP address
               Input format :
               xxx&xxx&xxx with IPVR = V4 or with IPVR = V6
                - mit xxx 1 to 3 digits [0..255]
               xx-xxx - mit xx 1 to 2 digits for the byte number
                        und xxx 1 to 3 digits for the byte value
     03 ippo - Remote IP port number
               Input format :
               xxxxxx - with xxxxx 1 to 5 digits
               Possible values : 0 .. 65535 */
               If the Remote IP Port =0, it is dynamically assigned */
     04 reru - Remote call number
               Input format:
               yxxxx - 5 digit line call number [y:A..F,x:0..9] */
/* "E" means parameter input */
rirr:<v4,192&168&13&121,5060;
```

Step | **Description 3.9.13** From TCO, administer settings as displayed. AOGD<lipi:? /* 01 lipp - Local IP Port Input format : xxxxxx - with xxxxx 1 to 5 digits Possible values : 0 .. 65535 If the Local IP Port =0, it is dynamically assigned */ /* 02 ipcm - Connectivity mode Input format : ${\tt xxxxxx}$ - with ${\tt xxxxx}$ 1 to 5 digits Possible values : 0 .. 65535 */ /* "E" means parameter input */ lipi:<5060,11; AOGD<aozu:s; AOGD<aozu:f; WARNING - No traffic group entered for service 19

Step | **Description 3.9.14** From TCO, verify the provisioning administered in this section as displayed. AOGD<anzg:c9000; ______ 26.04.07 13:50:38 Connecting circuit _____ Call No. : C9000 - C9029 Slot / HWA AO type : 01-01-07-30 : PRA ______ General ADS data Name : SIP_Trunk Accounting section : 00000 Protocols Protocol | Version | faulty | busy 2 | error ------ETSI | 60 | OFF | OFF | OFF Overload priority : 2
Public bar. unit gr. : 1
Colisee bar. unit gr. : 0
: 0 DISA-group Dealergroup
CN alloc. HKZ line & tie : -1 Dealergroup Category : -1
Waiting field maximum : 10 Reserved Reserved
Connection memory : 0
Service memory : 2
AO state : IN OPERATION
Service block : sv-free
Call number block : Off (V4)192.168.11.102:5060 Active Coder : g711alaw64k
Secured registration : NO
Remote IP - address : (V4)192.168.13.121:5060 Remote call number : Connectivity mode : 11 ______

From TCO, scroll down to continue verifying the provisioning administered in this section a displayed. TLP	Step	Description					
displayed. Service data	.9.15	From TCO, scroll	down to continu	e verifying the p	provisioning a	dministered in thi	is section as
TLP				, , ,			
TLP							
Status RELEASED RELEASED Dial group 2 7 Traffic group 1 0 Switchover group 0 0 Code dial group 0 0 LCR-group 0 0 Dial retrieval DEACTIVE DEACTIVE Backward rel. DEACTIVE DEACTIVE coder g711alaw64k g729annexa init init codermode normal normal		Service data					
Status RELEASED RELEASED Dial group 2 7 Traffic group 1 0 Switchover group 0 0 Code dial group 0 0 LCR-group 0 0 Dial retrieval DEACTIVE DEACTIVE Backward rel. DEACTIVE DEACTIVE coder g711alaw64k g729annexa init init codermode normal normal							
Dial group 2 7 Traffic group 1 0 Switchover group 0 0 Code dial group 0 0 LCR-group 0 0 Dial retrieval DEACTIVE DEACTIVE Backward rel. DEACTIVE DEACTIVE coder g711alaw64k g729annexa init codermode normal normal		ļ	TLP	GEN	Ţ	ļ	1
Traffic group 1 0 0 0 0 0 0 0 0 0 0		Status	RELEASED	RELEASED	-+ 	+ 	+
Switchover group 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0		Dial group	2	7			
Code dial group 0		Traffic group	1	0			
LCR-group 0 0 0 Dial retrieval DEACTIVE DEACTIVE DEACTIVE DEACTIVE Coder g711alaw64k g729annexa init codermode normal normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode codermode normal codermode co		Switchover group	0	0			
Dial retrieval DEACTIVE DEACTIVE Backward rel. DEACTIVE DEACTIVE Coder g711alaw64k g729annexa init init codermode normal normal codermode normal codermode normal codermode normal codermode normal codermode normal codermode codermode normal codermode codermode normal codermode cod		Code dial group	0	0			
Backward rel. DEACTIVE DEACTIVE coder g711alaw64k g729annexa init init codermode normal normal		LCR-group	0	0	İ		ĺ
coder g711alaw64k g729annexa init init codermode normal normal		Dial retrieval	DEACTIVE	DEACTIVE	İ		ĺ
codermode normal normal		Backward rel.	DEACTIVE	DEACTIVE	ĺ	İ	ĺ
		coder	g711alaw64k	g729annexa	init	init	İ
echocanceler on on		codermode	normal	normal	ĺ	į	İ
		echocanceler	on	on	j	İ	İ
		=========	=========	==========	=========	==========	====

Step | **Description 3.9.16** From TCO, scroll down to continue verifying the provisioning administered in this section as displayed. B channel data Allocation code : NSTA Deliberation code : DEACTIVE B chan. Bundle Direct Acc. Status | B chan. Bundle Direct Acc. Status number number number number 1 86 W O F 16 86 W O F 2 86 W O F 17 86 W O F 3 86 W O F 18 86 W O F 4 86 W O F 19 86 W O F 5 86 W O F 20 86 W O F 6 86 W O F 21 86 W O F 7 86 W O F 22 86 W O F 8 86 W O F 22 86 W O F 8 86 W O F 23 86 W O F 9 86 W O F 23 86 W O F 9 86 W O F 24 86 W O F 10 86 W O F 25 86 W O F 11 86 W O F 25 86 W O F 11 86 W O F 26 86 W O F 12 86 W O F 26 86 W O F 13 86 W O F 27 86 W O F 14 86 W O F 28 86 W O F 15 86 W O F 29 86 W O F Number of seizable B channels: 30 Seizure direction Status B - BUSY G - outgoing K - incoming D - DEFECT. W - bothway EB - EDSS1 BUSY ER - EDSS1 RESERVED F - FREE G - FAULTY R - RESERVED Access right ----- S - BARRED T - DEFECT./BARRED M - with O - without V - SEIZED/BARRED ______

3.10. Configure an Access Number for SIP Trunking

This section describes the steps for configuring an access number for SIP trunking on the Avaya I55. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step | **Description 3.10.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer settings to install an access code for SIP clients as follows: Enter 1:addr; at the PROL< command prompt. To query available options, enter a "?" at the **ADDR**< command prompt. To access the bundle addressing menu for bundle 86, enter **badm:86**; at the **ADDR**< command prompt. ______ CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS TC-03-3 03-05-2007 15:58:23 PROL (V4)192.168.12.144:1732 PROL<1:addr; Command processing in progress ! ADDR<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 pgwe - Program change */ /* 02 auge - Select output device for display */ /* 03 addm - branch to addressing menu */ /* 04 badm - branch to bundle addressing menu */ ADDR<badm:86;

3.10.2 From TCO, administer settings as displayed.

```
ADDR<?
/* SE Status display switch on */
/* HE Task-Help switch on */
/* GF Enable blocked device */
/* PE PC mode switch on */
/* 01 exit - Back to main menu */
/* 02 pdaz - Display Bundle Addressing Data */
/* 03 ipae - change Bundle ISDN-PABX-Access-Number */
/* 04 ppae - change Bundle Private-PABX-Access-Number */
/* 05 ipre - change Bundle ISDN-PABX-Prefixes */
/* 06 ppre - change Bundle Private-PABX-Prefixes */
/* 07 dead - change Bundle Destination Addresses */
/* 08 orad - change Bundle Origination Addresses */
/* 09 sads - set Bundle Address Settings */
/* 10 rads - reset Bundle Address Settings */
/* 11 ilev - change Bundle ISDN NP Level */
/* 12 plev - change Bundle Private NP Level */
/* 13 bnae - change bundle number */
ADDR<ipre:?
    01 rich - direction
           possible values: k - inward
                             g - outward */
     02 npre - NPI-Prefix
           max. 15 digits, possible values: [-1,0..9,*]
                      Init value = -1 */
     03 tpin - TON-Prefix international
           max. 15 digits, possible values: [-1,0..9,*]
                      Init value = -1 */
     04 tpna - TON-Prefix national
           max. 15 digits, possible values: [-1,0..9,*]
                      Init value = -1 */
     05 tpsu - TON-Prefix subscriber
           max. 15 digits, possible values: [-1,0..9,*]
                      Init value = -1 */
/* "E" means parameter input */
ipre:<g,0,00,0,;
ADDR<ilev:k,dddi;
```

Step | **Description 3.10.3** From TCO, verify the provisioning administered in this section as displayed. ADDR<pdaz; Data for Bundle 86 ______ - ISDN PABX Access numbers coun- nation. Access code def. DDI-Part try destin. subscriber subsr. number outward: - - - - - - - - -- PABX Access number: Priv.-NP Level2 Level1 Access Code def. DDI-Part code Local Local Nr outward: inward : -- Prefixes : ISDN-NP: outward : 0 inward : -inward : -- Level ISDN-NP outward : init inward : dddi <ilev - Level Priv.-NP outward : init inward : init Destination Addresses (Default) Internal Address : init Format of ISDN Address : init Format of Private Address : init Origination Addresses (Default) Internal Address : init Format of ISDN Address : init Format of Private Address : init Bundle Address Settings

3.11. Configure Supported Services for SIP Trunking

This section describes the steps for configuring supported services for SIP trunking on the Avaya I55. The configuration presented in this section is administered by means of MML commands entered in the TCO, which is accessible via ISM.

Step | **Description 3.11.1** From ISM, open TCO by selecting: **Service** → **TCO** (**Terminal emulation**). From TCO, administer settings to install an access code for SIP clients as follows: Enter 1:wabe; at the PROL< command prompt. To query available options, enter a "?" at the WABE< command prompt. CONSOLE-ID LOGIN DATE/TIME TASK CALL NUMBER/IP ADDRESS ______ TC-03-3 26-04-2007 14:07:46 PROL (V4)192.168.12.144:1378 PROL<1:wabe; Command processing in progress ! WABE<? /* SE Status display switch on */ /* HE Task-Help switch on */ /* GF Enable blocked device */ /* PE PC mode switch on */ /* 01 pgwe - Program change */ /* 02 auge - Select output device */ /* 03 anzg - Branch to display menu */ /* 04 akze - Set up AKZ */ /* 05 akzl - Clear AKZ */ /* 06 idke - Set up identification code */ /* 07 idkl - Clear identification code */ /* 08 cpnm - Install/delete company name */ /* 09 vbke - Set up VKZ */ /* 10 vbkl - Delete VKZ */ /* 11 datm - Branch to DAT DDI Alpha Tagging administration *//* 12 akza - Modify AKZ */

	ription											
Fron	n TCO, query s	ettings	admi	nistere	d in	Section	n 3.8 as displ	ayed				
WABE	<anzg;< th=""><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th></anzg;<>											
WABE	<dwgr:2,v;< th=""><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th><th></th></dwgr:2,v;<>											
								0	4.05.	07 1	3:49	:15
_	lay of dial e					_	-					
	======================================	===== 2	=====	=====	====	=====	===					
	l method : 1	_	1									
Dia	i meerioa - 1	rrara	_									
AKZ	Dial	Bndl	AKZ	SA	Co.	LCR	dialing		ext.	LCR	RI-	Num.
	sele.	numb.	Info	group	nr.	data	conversion		all.	rout	SA	Plan
						set	digits	sel	cat.	flg	flg	
		_						_				
0	EXTERN	3	-	-	-	-		0 -		ROFF		
1	NEED	1.0	1						xch.			eartn
1	NETZ	12	1	-	-	_		0		ROFF		-
2	NETZ	12 12	1 1	-	-	_		0		ROFF		-
3 4	NETZ INTERN		⊥ 2	_	-	_		0 0	TNT.I.	ROFF	_	-
5	NETZ		1	_		_		0	_ TNTTT	ROFF	_	_
6	NETZ		1	_	_	_		0		ROFF		_
7	NETZ		1	_	_	_		0		ROFF		_
81	NETZ	12	1	_				0		ROFF		
82	INTERN		5	_	_	_		0	TIVT 1	- KOFF	_	_
991	INTERN		2	_	_	_		0	_	_	_	_
A	RUVA	_	_	_	_	_	_	_	_	_	_	_
C	PUALLG		_	_	_	_	_	_	_	_	_	_
D	CW	_	_	_	_	_	_	_	_	_	_	_
E0	EXTERN		3	_	_	_		0	INIT	ROFF	_	_
			_					_				

Step | **Description 3.11.3** From TCO, administer settings as displayed. WABE<1; WABE<1:aolm; Command processing in progress ! AOLM<3:c9000; AOLM<4; 04.05.07 13:51:27 AO-Number AO - Perform. features (Service: TLP) C9000 NAM CWA AOLM<falm:amt; AOLM<falm:toc,klf; AOLM<4; 04.05.07 13:52:00 AO-Number AO - Perform. features (Service: TLP) C9000 AMT CWA KLF TOC AOLM<salm:cwa; AOLM<4; 04.05.07 13:55:39 AO-Number AO - Perform. features (Service: TLP) C9000 AMT KLF TOC AOLM<1; AOLM<1:anlm; Command processing in progress ! ANLM<9;

```
Step | Description
3.11.4 From TCO, administer settings as displayed.
       ANLM<?
       /* SE Status display switch on */
       /* HE Task-Help switch on */
       /* GF Enable blocked device */
       /* PE PC mode switch on */
       /* 01 pgwe - Program change */
       /* 02 auge - Select output device */
/* 03 almf - Enable System LM */
       /* 04 alms - Blocking System LM */
       /* 05 alma - Display System LM */
       /* 06 alvf - Enable System LM variants */
       /* 07 alvs - Blocking System LM variants */
       /* 08 alva - Display System LM variants to
                   one System LM */
       /* 09 lmab - Display perf. feature capacity of system */
       ANLM<almf:toc,klf;
       ANLM<alma:toc;
        TOC F
       ANLM<alma:klf;
        KLF F
```

4. Avaya Meeting Exchange Configuration

This section displays the configuration for enabling Avaya Meeting Exchange to interoperate with the Avaya I55. Avaya Meeting Exchange is administered and maintained using a standard web browser over a secure connection by entering

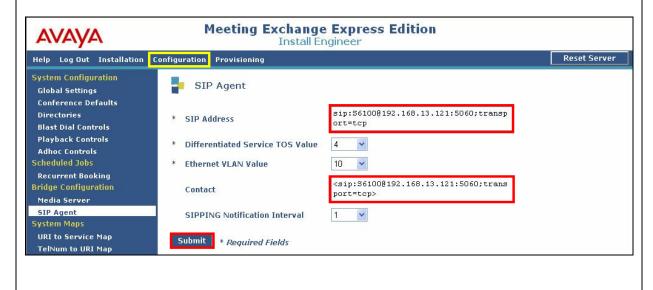
https://<IP address of Avaya Meeting Exchange>/mx into the web browser's Uniform Resource Locator (URL) bar.

4.1. Configure Connectivity

This section describes the steps for configuring SIP connectivity between Avaya Meeting Exchange and other SIP User Agents (UA).

Step | **Description**

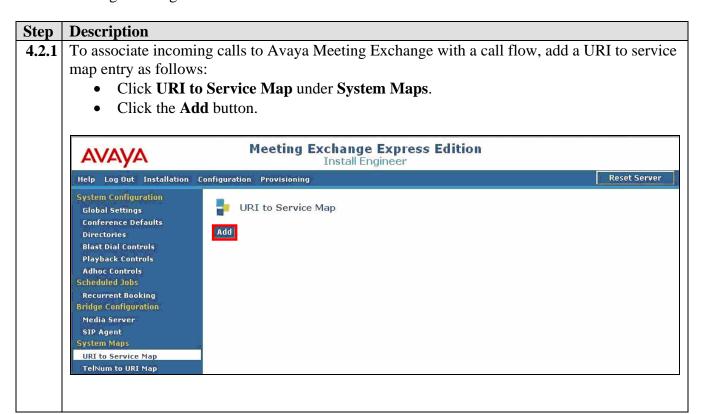
- **4.1.1** Administer settings that enable SIP connectivity between Avaya Meeting Exchange and the Avaya I55 as follows:
 - From the web interface toolbar, click **Configuration**.
 - Click **SIP Agent** under **Bridge Configuration**.
 - Enter a SIP URI for Avaya Meeting Exchange that conforms to SIP standards in the SIP Address field. This field is used to populate the From Header Field in SIP INVITE messages from Avaya Meeting Exchange. To enable SIP/TCP connectivity on port 5060, this entry must contain 5060 and transport=tcp. The user field, S6100, must conform to SIP standards, and is selected to uniquely identify this server. For example, S6100 will be inserted in the From Header Field of SIP INVITE messages from Avaya Meeting Exchange and will display on a participant's endpoint when Dial-Out procedures from Avaya Meeting Exchange are invoked. This enables an end-user to identify a call from Avaya Meeting Exchange.
 - Enter the SIP URI, as configured for the **SIP Address** field, in angled brackets in the **Contact** field. This field is used to populate the Contact Header Field in SIP INVITE messages from Avaya Meeting Exchange, and provides SIP UAs, for these Application Notes the Avaya I55, a means for acknowledging SIP messages from Avaya Meeting Exchange.
 - Use default settings for remaining fields.
 - Click the **Submit** button to add the configuration to the database.



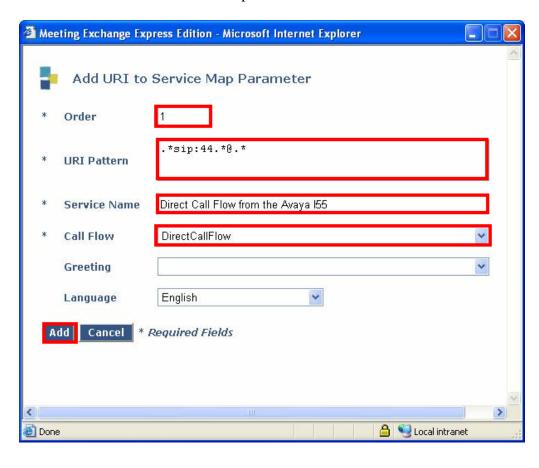
4.2. Configure Call Routing

This section describes the steps for configuring call routing for Avaya Meeting Exchange. On Avaya Meeting Exchange, call routing is defined by service maps as follows:

- For inbound calls to Avaya Meeting Exchange, service maps for URI to telephone number translations are utilized. These translations associate calls to Avaya Meeting Exchange with corresponding call flows, thus allowing for specific treatment for an incoming call based on a the caller's SIP Uniform Resource Identifier (URI).
- For outbound calls from Avaya Meeting Exchange, service maps for telephone number to URI translations are utilized. These translations associate a telephone number pattern with a corresponding SIP URI of a SIP UA, thus allowing call origination from Avaya Meeting Exchange to the SIP UA.

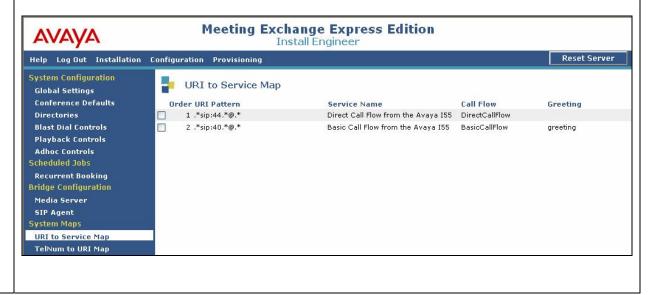


- **4.2.2** From the **Add URI to Service Map Parameter** screen, administer settings to enable a direct call flow for calls from the Avaya I55 as follows:
 - Leave the **Order** field at the default value. Avaya Meeting Exchange parses URI to service map entries for pattern matches in descending order, terminating the search once a pattern is matched. For this sample configuration, order is irrelevant as the patterns for call flows are mutually exclusive.
 - Enter a rule in the **URI Pattern** field to match the pattern of incoming Request URIs in SIP INVITE messages from the Avaya I55. Metacharacters such as "." (matches any one character) or "*" (matches zero or more of the preceding character) may be utilized. For example, assume the Avaya I55 sends the following URI: sip:444@192.168.13.121:5060;transport=tcp. The entry in the **URI Pattern** field, .*sip:44.*@.*, would match sip:44, then zero or more characters, followed by @, then zero or more characters.
 - To allow access to conferences as moderator, without entering a passcode, select **DirectCallFlow** from the drop down menu for the **Call Flow** field.
 - Enter a descriptive name for this map in the **Service Name** field.
 - Click the **Add** button to add the map to the database.



- **4.2.3** To associate incoming calls to Avaya Meeting Exchange with a basic call flow, repeat **Step 4.2.1** to add a URI to service map entry for a basic call flow with the following parameters:
 - Leave the **Order** field at the default value.
 - Enter .*sip:40.*@.* in the URI Pattern field to match the pattern of incoming Request URIs in SIP INVITE messages from the Avaya I55.
 - To access a conference with an associated passcode, select **BasicCallFlow** from the drop down menu for the **Call Flow** field.
 - Enter a descriptive name for this map in the **Service Name** field.
 - [Optional] Select a greeting from the drop down menu for the Greeting field.
 - The resulting URI to service map list is displayed below.

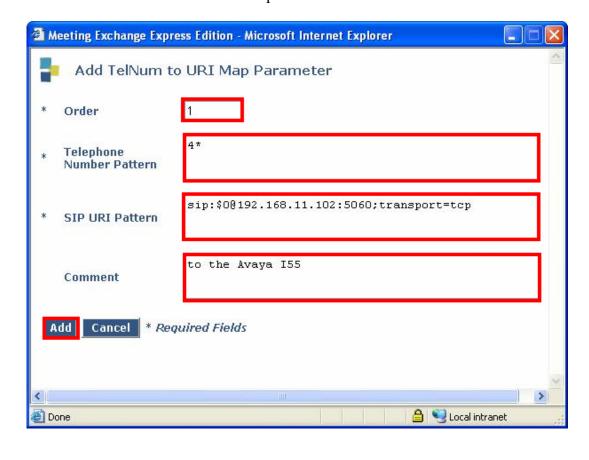
Note: The provisioning for the **URI Pattern** fields for the direct and basic call flows utilize wild cards that make the call flows mutually exclusive while maximizing the breadth of the pattern match. For example, the **URI Pattern** field for the basic call flow is .*sip:40.*@.*. This aligns with the provisioning for call routing on the Avaya I55, and allows 40x, where x can be any digit, to match this direct call flow.

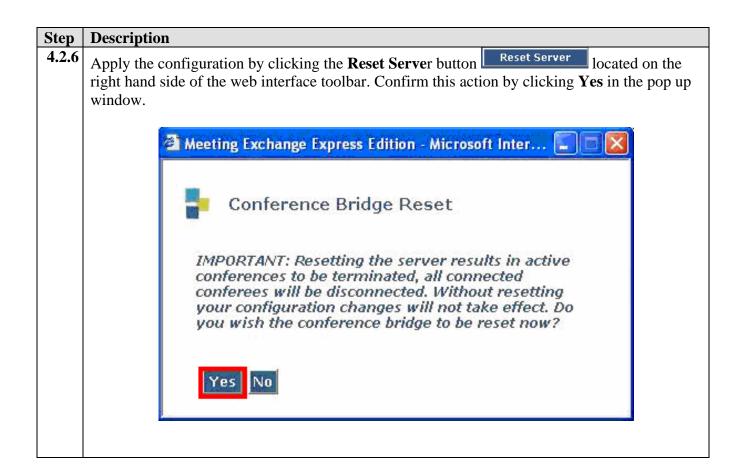


Description Step 4.2.4 To enable routing of outbound calls from Avaya Meeting Exchange, add a TelNum to URI map entry as follows: Click TelNum to URI Map under System Maps. Click the **Add** button. **Meeting Exchange Express Edition** AVAVA **Install Engineer** Reset Server Help Log Out Installation Configuration Provisioning TelNum to URI Map **Global Settings** Conference Defaults Directories Blast Dial Controls **Playback Controls** Adhoc Controls Recurrent Booking Media Server SIP Agent

URI to Service Map TelNum to URI Map

- **4.2.5** From the **Add TelNum to URI Map Parameter** screen, administer settings to enable outbound calling to the Avaya I55 as follows:
 - Leave the **Order** field at the default value. Avaya Meeting Exchange parses TelNum to URI map entries for pattern matches in descending order, terminating the search once a pattern is matched. For this sample configuration, order is irrelevant as there is only one entry in the database.
 - Enter a rule in the **Telephone Number Pattern** field that matches the administration for telephone extensions on the Avaya I55. Metacharacters such as "*" (refers to a character string) or "?" (refers to a single character) may be utilized.
 - To enable outbound calling from Avaya Meeting Exchange, enter a rule in the **SIP URI Pattern** field that conforms to SIP standards. To enable SIP/TCP connectivity for outbound calls to the Avaya I55, the rule must contain **5060** and **transport=tcp**. The metacharacter, **\$0** is replaced by the entire **Telephone Number Pattern** at the location of **\$0** in the **SIP URI Pattern**. For example, if **401** is the dialed string, Avaya Meeting Exchange will send a SIP INVITE message with a SIP URI and To Header Field formatted as follows:
 - sip:401@192.168.11.102:5060;transport=tcp.
 - Click the **Add** button to add the map to the database.





4.3. Provision Accounts

The following steps present an example of provisioning an end user account and associated conference reservation on Avaya Meeting Exchange.

Step **Description** To provide end users access to the conferencing features available on Avaya Meeting 4.3.1 Exchange, add an end user account as follows: From the web interface toolbar, click **Provisioning**. Click End User Accounts under Provisioning. Click the **Add** button. *Note*: Avaya Meeting Exchange comes with pre-provisioned accounts as displayed. Meeting Exchange Express Edition AVAVA Install Engineer Reset Server Help Log Out Installation Configuration Provisioning **End User Accounts** My Account **Conference Reservations** Name E-Mail Administrator Accounts **End User Accounts** Phone Enabled Enabled & Disabled **Bulk Upload Scheduling** Search Server Configuration Schedulina Enabled Status Phone Number CSV Account 0 1234556660 csv@account0.com CSV Account 1 csv@account1.com 1234556661 CSV Account 2 csv@account2.com 1234556662 CSV Account 3 1234556663 csv@account3.com

csv@account4.com

CSV Account 4

CSV Account 5

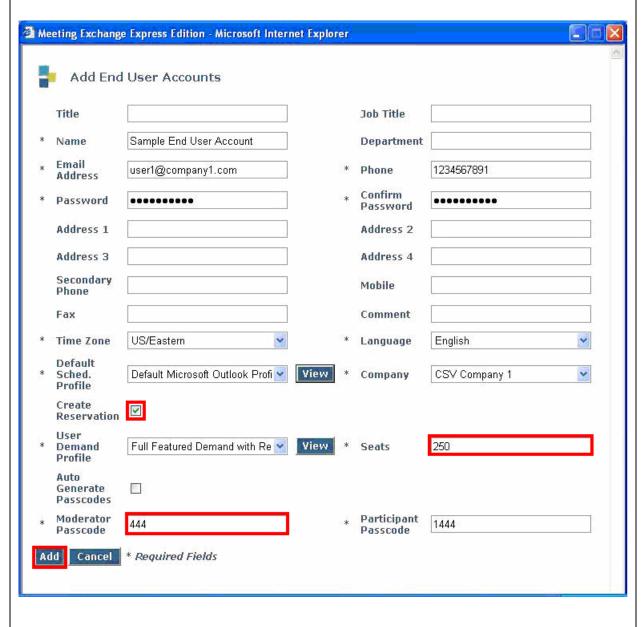
Add Edit Disable

< Page 1 of 1 > >> Total: 6 Rows/Page: 10

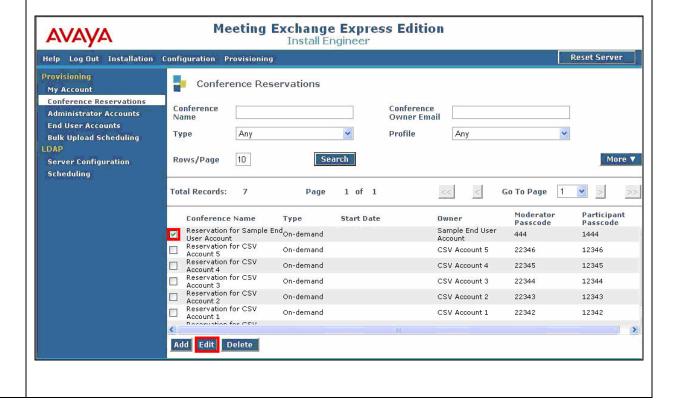
1234556664

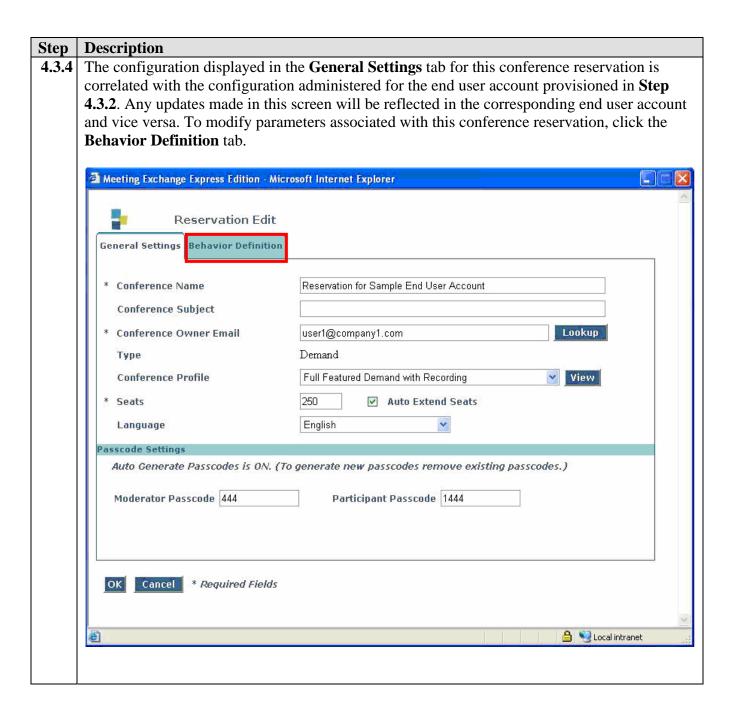
1234556665

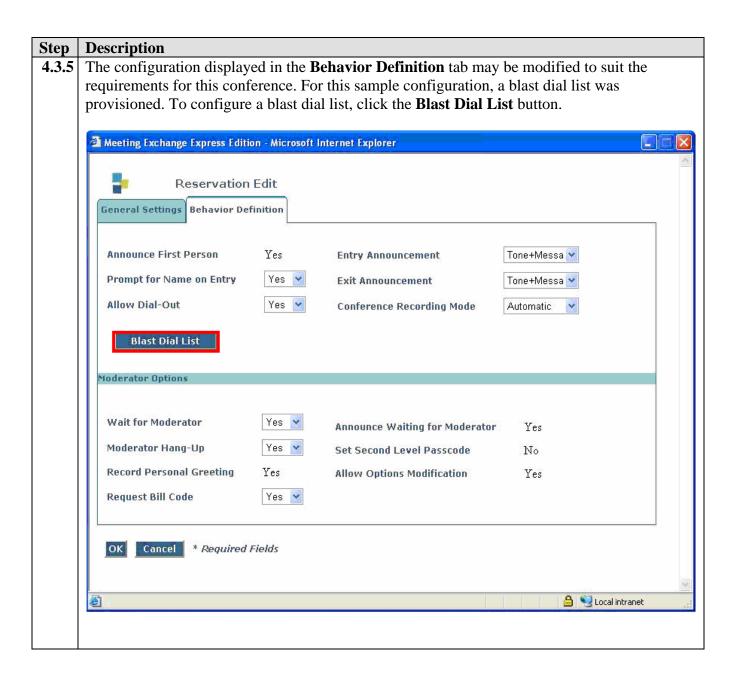
- **4.3.2** From the **Add End User Accounts** screen, provision an end user account as follows:
 - Check **Create Reservation** to generate a reservation for an on demand conference that is associated with this end user account.
 - Enter the number of ports assigned to this conference in the **Seats** field.
 - Enter a number in the **Moderator Passcode** field that corresponds to the direct call flow provisioned in **Step 4.2.2**.
 - Refer to [1] for definitions regarding the remaining required fields on this screen.
 - Click the **Add** button to add the account to the database.



- **4.3.3** Modify the conference reservation corresponding to the end user account provisioned in **Step 4.3.2** as follows:
 - Click Conference Reservations under Provisioning.
 - Check the conference reservation corresponding to the end user account provisioned in **Step 4.3.2**.
 - Click the **Edit** button.







Description Step **4.3.6** From the **Blast Dial List** screen, add entries to the blast dial list as follows: Enter a number in the **Phone Number** field that corresponds to an extension of an Avaya 4600 Series IP Telephone that is registered to the Avaya I55. Enter a descriptive name for this phone number in the **Name** field. Click the **Add** button to add entries to this blast dial list. 🚰 Meeting Exchange Express Edition - Microsoft Internet Explorer **Blast Dial List** Phone Number Avaγa-l55-401 Name Add **Blast Dial Users** Save Cancel

Description Step Repeat Step 4.3.6 to add additional phone numbers to the blast dial list. The resultant blast dial 4.3.7 list is displayed below. • Click the **Save** button to save and associate the blast dial list with this conference. • Click the **OK** button (displayed in the lower left hand corner of the **Behavior Definition** tab in **Step 4.3.5**) to save the modifications to this conference in the database. 🚰 Meeting Exchange Express Edition - Microsoft Internet Explorer **Blast Dial List** Phone Number Name Add **Blast Dial Users** Name **Phone Number** 401 Avaya-I55-401 402 Avaya-155-402 403 Avaya-155-403 404 Avaya-155-404 405 Avaya-155-405 Cancel Delete Save

5. Interoperability Compliance Testing

5.1. General Test Approach

The general test approach was to place calls between the Avaya I55 and Avaya Meeting Exchange, utilizing the sample configuration displayed in **Figure 1**.

The main objectives were to verify the following:

- Inbound calling from Avaya 4600 Series IP Telephones registered to the Avaya I55 to scheduled and demand conferences provisioned on Avaya Meeting Exchange:
 - o Direct call flow (without participant-access-code)
 - o Basic call flow (with participant-access-code)
- Outbound calling from Avaya Meeting Exchange to Avaya 4600 Series IP Telephones registered to the Avaya I55:
 - o Blast dial to a pre-provisioned blast dial list
 - o Originator dial-out
- Conferencing features, provided by Avaya Meeting Exchange, for both moderator and participant accessed during a conference call via touchtone commands.
- The following sub-set of the SIPPING-19 supplementary features for SIP endpoints:
 - o Call hold
 - Attended/unattended call transfer
 - o Call forward
 - o Three-way conference
- The following transport methods for signaling between the Avaya I55 and Avaya Meeting Exchange:
 - o TCP
 - o UDP
- The following codecs:
 - o G711MU
- Subjective voice quality for endpoints participating in a conference.
- DTMF transmission via RFC 2833.

5.2. Test Results

All test cases, as defined by the general test approach, passed.

6. Verification Steps

The following steps were used to verify the administrative steps presented in these Application Notes and are applicable for similar configurations in the field. The verification steps in this section validated the following:

Step	Description
6.1.1	Validate signaling and media connectivity for inbound calls to Avaya Meeting Exchange from the Avaya I55.
	• From a station registered to the Avaya I55, dial 444 to enter the conference provisioned in Section 4.3 as moderator via the direct call flow provisioned in Step 4.2.2 .
6.1.2	 Validate signaling and media connectivity for outbound calls from Avaya Meeting Exchange to the Avaya I55. From a station in a conference on Avaya Meeting Exchange, enter the appropriate touchtone command to invoke a blast dial to the blast dial list provisioned in Section 4.3.
6.1.3	 Verify that calls to and from Avaya Meeting Exchange are managed correctly, e.g., callers are added/removed from conferences. This is verified by the following procedures: Log in to the Avaya Meeting Exchange server console with the appropriate credentials. At the command prompt, enter the command: watch -t -n 5 -d ''ipinfo -l egrep -ci active'' This command provides a real time, continuous update of port utilization on Avaya Meeting Exchange.
6.1.4	The following packet trace depicts the inbound and outbound call scenarios invoked in Step 6.1.1 and Step 6.1.2 . Time 192,168,11,102 192,168,13,121 Comment
	13.549 13.550 13.550 13.553 13.553 13.557 13.557 13.558 13.557 13.559 13

7. Conclusion

These Application Notes present a compliance-tested solution comprised of the Avaya I55 and Avaya Meeting Exchange. This solution enables connectivity between the Avaya I55 and Avaya Meeting Exchange utilizing standards based SIP connectivity.

8. Additional References

Avaya references are available at http://support.avaya.com.

- [1] Avaya Meeting Exchange Express Edition Release 1.5 Administration and Maintenance Guide, Issue 1, Doc ID: 04-601909, March 2007.
- [2] Avaya Meeting Exchange Express Edition Release 1.5 Installation and Configuration Guide, Issue 1, Doc ID: 04-601898, March 2007.

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