Abstract

These Application Notes present the procedures for configuring connectivity between Avaya Communication Manager, Avaya Meeting Exchange Express Edition (Avaya Meeting Exchange), and the Cantata Technology IMG 1010 Media Gateway (IMG). The IMG provided T1 CAS to SIP gateway functionality between Avaya Communication Manager and Avaya Meeting Exchange. This configuration enables telephones registered to either Avaya Communication Manager, or Avaya SIP Enablement Services access to a rich set of audio conferencing options provided by Avaya Meeting Exchange via the IMG.

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 present the procedures for configuring connectivity between Avaya Communication Manager, Avaya Meeting Exchange Express Edition (Avaya Meeting Exchange), and the Cantata Technology IMG 1010 Media Gateway (IMG). The IMG provided T1 CAS to SIP gateway functionality between Avaya Communication Manager and Avaya Meeting Exchange. This configuration enables telephones registered to either Avaya Communication Manager, or Avaya SIP Enablement Services access to a rich set of audio conferencing options provided by Avaya Meeting Exchange via the IMG.

Figure 1 illustrates the sample configuration utilized for this compliance tested solution. Avaya Communication Manager provided endpoint aggregation and media gateway functionality. For example, any telephone or trunk type associated with Avaya Communication Manager can interoperate with Avaya Meeting Exchange via the IMG. For this sample configuration, SIP, H.323, Digital, and Analog telephones were utilized.

Avaya Meeting Exchange is a SIP-based voice conferencing solution that runs on an S6100 server and 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 Avaya Communication Manager 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 either Avaya Communication Manager, or Avaya SIP Enablement Services.

The IMG provides network connectivity for voice services, enabling the delivery of VoIP services via SIP into ISDN-PRI, CAS and SS7 networks, as well as IP to IP transcoding for network peering applications. For this sample configuration, the IMG provided SIP connectivity to Avaya Meeting Exchange and T1 CAS connectivity to Avaya Communication Manager.
The end-to-end signaling and media connectivity is as follows:

- Signaling (SIP) and media (RTP) connectivity between Avaya Meeting Exchange and the IMG is depicted by the green dashed line.
- T1 signaling and media (CAS) connectivity between Avaya Communication Manager and the IMG is depicted by the blue dotted line.

To account for the SIP telephones in this sample configuration, Avaya SIP Enablement Services was utilized as a SIP registration server only.

Figure 1: Sample Configuration
2. Equipment and Software Validated

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

<table>
<thead>
<tr>
<th>Equipment</th>
<th>Software Version</th>
</tr>
</thead>
<tbody>
<tr>
<td>Avaya S8710 Servers</td>
<td>Avaya Communication Manager 4.0 (R014x.00.1.731.2)</td>
</tr>
<tr>
<td>Avaya G650 Media Gateway</td>
<td></td>
</tr>
<tr>
<td>• Avaya TN2312BP (IPSI)</td>
<td>HW12 FW040</td>
</tr>
<tr>
<td>• Avaya TN799DP (C-LAN)</td>
<td>HW01 FW024</td>
</tr>
<tr>
<td>• Avaya TN2302AP (MEDPRO)</td>
<td>HW20 FW117</td>
</tr>
<tr>
<td>Avaya Meeting Exchange Express Edition</td>
<td>S6100-2.5.60.0</td>
</tr>
<tr>
<td>Avaya SIP Enablement Services</td>
<td>SES04.0-04.0.033.6</td>
</tr>
<tr>
<td>Avaya C364T-PWR Converged Stackable Switch</td>
<td>4.5.14</td>
</tr>
<tr>
<td>Avaya 4600 Series IP Telephones</td>
<td>2.8 (H.323)</td>
</tr>
<tr>
<td>Avaya 4600 Series IP Telephones</td>
<td>2.2.2 (SIP)</td>
</tr>
<tr>
<td>Avaya 6408D+ Digital Telephones</td>
<td>--</td>
</tr>
<tr>
<td>Analog Telephones</td>
<td>--</td>
</tr>
<tr>
<td>Cantata Technology IMG 1010 Media Gateway</td>
<td>10.3.3</td>
</tr>
<tr>
<td>Cantata Technology IMG 1010 GateControl Element</td>
<td></td>
</tr>
<tr>
<td>Management System</td>
<td>10.3.3.174</td>
</tr>
<tr>
<td>Cantata Technology ClientView</td>
<td>10.3.3.174</td>
</tr>
</tbody>
</table>

Table 1: Equipment and Software Versions
3. Avaya Communication Manager Configuration

This section displays the configuration for enabling Avaya Communication Manager to interoperate with Avaya Meeting Exchange via the IMG.

Avaya Communication Manager was administered from the System Access Terminal (SAT). In these Application Notes the SAT screens are shown with a gray shaded background. In some instances, the information from the original screen has been edited or annotated for brevity or clarity in presentation. For example, entries and/or fields in the SAT screens that were either modified or were required for these Application Notes are displayed with boldface type. Refer to [3] and [4] for additional information regarding the configuration displayed in this section.

3.1. Verify Licensing

The following steps verify licensing on Avaya Communication Manager that is required to support the configuration displayed in these Application Notes. If a required feature is not enabled or there is insufficient capacity, contact an authorized Avaya account representative to make the appropriate changes.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>3.1.1</td>
<td>Issue the command “display system-parameters customer-options”, and proceed to page 3. Verify that the <strong>ARS/AAR Dialing without FAC</strong> field is enabled.</td>
</tr>
</tbody>
</table>

*Note: The **ARS/AAR Dialing without FAC** feature allows direct access to Automatic Alternate Routing (AAR) and Automatic Route Selection (ARS) from the dial plan analysis table.*

```
<table>
<thead>
<tr>
<th>OPTIONAL FEATURES</th>
</tr>
</thead>
<tbody>
<tr>
<td>Access Security Gateway (ASG)? n</td>
</tr>
<tr>
<td>Analog Trunk Incoming Call ID? n</td>
</tr>
<tr>
<td>A/D Grp/Sys List Dialing Start at 01? n</td>
</tr>
<tr>
<td>Answer Supervision by Call Classifier? n</td>
</tr>
<tr>
<td>ARS? y</td>
</tr>
<tr>
<td>ARS/AAR Partitioning? y</td>
</tr>
<tr>
<td><strong>ARS/AAR Dialing without FAC? y</strong></td>
</tr>
<tr>
<td>ASA1 Link Core Capabilities? n</td>
</tr>
<tr>
<td>ASA1 Link Plus Capabilities? n</td>
</tr>
<tr>
<td>Async. Transfer Mode (ATM) PNC? n</td>
</tr>
<tr>
<td>Async. Transfer Mode (ATM) Trunking? n</td>
</tr>
<tr>
<td>ATM WAN Spare Processor? n</td>
</tr>
<tr>
<td>ATM8? n</td>
</tr>
<tr>
<td>Attendant Vectoring? y</td>
</tr>
</tbody>
</table>

(Note: You must logoff & login to effect the permission changes.)
```
### 3.2. Configure Connectivity

This section describes the steps for configuring CAS trunking between Avaya Communication Manager and the IMG.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 3.2.1 | Issue the command “**add ds1 <xxxxx>**”, where *xxxxx* is the location of the DS1 circuit pack in the Avaya G650 Media Gateway, and administer settings as displayed.  
- Enter a descriptive name for the DS1 circuit pack in the **Name** field.  
- Set the **Signaling Mode** field to **robbed-bit**.  
- Configure additional fields with boldface type as displayed, and use default settings for remaining fields. |

```plaintext
add ds1 1a07

<table>
<thead>
<tr>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Location: 01A07</td>
</tr>
<tr>
<td>Bit Rate: 1.544</td>
</tr>
<tr>
<td>Line Compensation: 1</td>
</tr>
</tbody>
</table>

**Signaling Mode**: robbed-bit

**Interface Companding**: mulaw

**Idle Code**: 11111111

**Slip Detection?**: n

**Near-end CSU Type**: other
```
3.2.2 Issue the command “add trunk-group <n>”, where <n> is the number of an unallocated trunk group, and administer settings as displayed.

- Enter a descriptive name for the trunk group in the Name field.
- Set the Group Type field to tie.
- Enter a number in the TAC (Trunk Access Code) field that is consistent with the configuration for the dial plan.
- Set the Trunk Type field to a value that is compatible with the IMG media gateway settings.
- Configure additional fields with boldface type as displayed, and use default settings for remaining fields.

```
add trunk-group 7
TRUNK GROUP

Group Number: 7
Group Name: CAS Trunk to IMG-1010
Direction: two-way
Dial Access? y
Queue Length: 0
Comm Type: voice
Trunk Type (in/out): wink/wink

Group Type: tie
CDR Reports: y
COR: 1
TN: 1
TAC: 107
Outgoing Display? y
Busy Threshold: 255
Incoming Destination:

Trunk Flash? n
Auth Code? n
```

Add trunk-group 7
### Step 3.2.3
Proceed to Page 5, and administer the members for the trunk group as displayed.
- Enter `xxxxxyy` in the **Port** field, where `xxxx` corresponds the location of the DS1 circuit pack in the Avaya G650 Media Gateway, and `yy` corresponds to the trunk group member.

<table>
<thead>
<tr>
<th>Port</th>
<th>Code</th>
<th>Sfx</th>
<th>Name</th>
<th>Night</th>
<th>Mode</th>
<th>Type</th>
<th>Ans Delay</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>01A0701</td>
<td>TN464</td>
<td>F</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
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<tr>
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</tr>
<tr>
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<td></td>
<td></td>
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<td></td>
</tr>
</tbody>
</table>
3.3. Configure Call Routing

This section describes the steps for configuring call routing from Avaya Communication Manager to Avaya Meeting Exchange via the IMG. For this sample configuration, ARS/AAR dialing without FAC is utilized to route calls to Avaya Meeting Exchange. Note that other forms of call routing may be utilized.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>3.3.1</td>
<td>Issue the command “<code>change dialplan analysis</code>”, and administer settings to route any numbers beginning with a 4 and totaling 3 digits in length via AAR as displayed.</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>change dialplan analysis</th>
<th></th>
</tr>
</thead>
<tbody>
<tr>
<td>DIAL PLAN ANALYSIS TABLE</td>
<td>Page 1 of 12</td>
</tr>
<tr>
<td>Percent Full: 1</td>
<td></td>
</tr>
<tr>
<td>Dialed String</td>
<td>Total Length</td>
</tr>
<tr>
<td>0</td>
<td>1</td>
</tr>
<tr>
<td>1</td>
<td>3</td>
</tr>
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<td>2</td>
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<td>2</td>
</tr>
<tr>
<td>9</td>
<td>2</td>
</tr>
<tr>
<td>*</td>
<td>1</td>
</tr>
<tr>
<td>#</td>
<td>3</td>
</tr>
</tbody>
</table>
### Step 3.3.2

Issue the command "**change route-pattern <n>**", where **n** is the number of an unallocated route pattern. Administer settings to utilize the trunk group provisioned in **Step 3.2.2** to route calls from Avaya Communication Manager to the IMG.

- Enter the number of the trunk group that was provisioned in **Step 3.2.2** in the **Grp No** field.
- To disable restrictions for call routing via this route pattern, set the Facility Restriction Level (**FRL**) field to the lowest setting.
- Configure additional fields with boldface type as displayed, and use default settings for remaining fields.

#### change route-pattern 7

<table>
<thead>
<tr>
<th>Pattern Number: 7</th>
<th>Pattern Name: CAS Rt To IMG</th>
</tr>
</thead>
<tbody>
<tr>
<td>Grp No FRL NPA Pfx Hop Toll No. Inserted</td>
<td>DCS/ IXC</td>
</tr>
<tr>
<td>1: 7 0</td>
<td>n user</td>
</tr>
<tr>
<td>2:</td>
<td>n user</td>
</tr>
<tr>
<td>3:</td>
<td>n user</td>
</tr>
<tr>
<td>4:</td>
<td>n user</td>
</tr>
<tr>
<td>5:</td>
<td>n user</td>
</tr>
<tr>
<td>6:</td>
<td>n user</td>
</tr>
</tbody>
</table>

#### BCC VALUE TSC CA-TSC ITC BCIE Service/Feature PARM No. Numbering LAR Dgts Format Subaddress

| 1: | y y y y y n n | rest | none |
| 2: | y y y y y n n | rest | none |
| 3: | y y y y y n n | rest | none |
| 4: | y y y y y n n | rest | none |
| 5: | y y y y y n n | rest | none |
| 6: | y y y y y n n | rest | none |
### Step 3.3.3

Issue the command “change aar analysis x” and add an entry in the table to utilize the route pattern provisioned in Step 3.3.2.

- Enter a number in the **Dialed String** field that will be utilized by Avaya Meeting Exchange to map to a direct call flow.
- Enter the number of the route pattern provisioned in Step 3.3.2 in the **Route Pattern** field.
- Configure additional fields with boldface type as displayed, and use default settings for remaining fields.

<table>
<thead>
<tr>
<th>Dialed String</th>
<th>Total Min</th>
<th>Total Max</th>
<th>Route Pattern</th>
<th>Call Type</th>
<th>Node Num</th>
<th>Call Type ANI</th>
<th>Route Pattern ANI</th>
<th>Call Type ANI Reqd</th>
</tr>
</thead>
<tbody>
<tr>
<td>401</td>
<td>3</td>
<td>3</td>
<td>7</td>
<td>aar</td>
<td>n</td>
<td>aar</td>
<td>n</td>
<td>n</td>
</tr>
<tr>
<td>444</td>
<td>3</td>
<td>3</td>
<td>7</td>
<td>aar</td>
<td>n</td>
<td>aar</td>
<td>n</td>
<td>n</td>
</tr>
</tbody>
</table>

### 4. Avaya Meeting Exchange Configuration

This section displays the configuration for enabling Avaya Meeting Exchange to interoperate with Avaya Communication Manager via the IMG. 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/TCP connectivity between Avaya Meeting Exchange and the IMG.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>4.1.1</td>
<td>Administer settings that enable SIP connectivity between Avaya Meeting Exchange and other SIP User Agents as follows:</td>
</tr>
<tr>
<td></td>
<td>• From the web interface toolbar, click <strong>Configuration</strong>.</td>
</tr>
<tr>
<td></td>
<td>• Click <strong>SIP Agent</strong> under <strong>Bridge Configuration</strong>.</td>
</tr>
<tr>
<td></td>
<td>• Enter a SIP URI for Avaya Meeting Exchange that conforms to SIP standards in the <strong>SIP Address</strong> 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 <strong>5060</strong> and <strong>transport=tcp</strong>. The user field, <strong>S6100</strong>, must conform to SIP standards, and is selected to uniquely identify this server. For example, <strong>S6100</strong> 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 allows end-user’s to identify a call from Avaya Meeting Exchange.</td>
</tr>
<tr>
<td></td>
<td>• Enter the SIP URI, as configured for the <strong>SIP Address</strong> field, in angled brackets in the <strong>Contact</strong> field. This field is used to populate the Contact Header Field in SIP INVITE messages from Avaya Meeting Exchange, and provides SIP User Agents, for these Application Notes the IMG, a means for acknowledging SIP messages from Avaya Meeting Exchange.</td>
</tr>
<tr>
<td></td>
<td>• Use default settings for remaining fields.</td>
</tr>
<tr>
<td></td>
<td>• Click the <strong>Submit</strong> button to add the configuration to the database.</td>
</tr>
</tbody>
</table>
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 a participant based on incoming calls based on a 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 User Agent (UA), thus allowing call origination from Avaya Meeting Exchange to the SIP UA.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 4.2.1 | To associate incoming calls to Avaya Meeting Exchange with a call flow, add a URI to service map entry as follows:  
  - Click **URI to Service Map** under System Maps.  
  - Click the **Add** button. |
4.2.2 From the **Add URI to Service Map Parameter** screen, administer settings to enable a direct call flow for calls from Avaya Communication Manager via the IMG 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 Avaya Communication Manager via the IMG. Metacharacters such as . (matches any one character) or * (matches zero or more of the preceding character) may be utilized. For example, assume the IMG 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 URLs in SIP INVITE messages from Avaya Communication Manager via the IMG.
- 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.
- 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 Avaya Communication Manager in Section 3.3, and allows 40x, where x can be any digit, to match this direct call flow.
<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 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. |
<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 4.2.5 | From the **Add TelNum to URI Map Parameter** screen, administer settings to enable outbound calling to Avaya Communication Manager via the IMG 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 on for telephone extensions on Avaya Communication Manager. 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 Avaya Communication Manager via the IMG, 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 31002 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:31002@192.168.13.112:5060;transport=tcp`.  
  - Click the **Add** button to add the map to the database. |
Apply the configuration by clicking the **Reset Server** button located on the right hand side of the web interface toolbar. Confirm this action by clicking **Yes** in the pop up window.
4.3. Provision Accounts

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

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>4.3.1</td>
<td>To provide end users access to the conferencing features available on Avaya Meeting Exchange, add an end user account as follows:</td>
</tr>
<tr>
<td></td>
<td>• From the web interface toolbar, click <strong>Provisioning</strong>.</td>
</tr>
<tr>
<td></td>
<td>• Click <strong>End User Accounts</strong> under <strong>Provisioning</strong>.</td>
</tr>
<tr>
<td></td>
<td>• Click the <strong>Add</strong> button.</td>
</tr>
</tbody>
</table>

**Note:** Avaya Meeting Exchange comes with pre-provisioned accounts as displayed.

![Meeting Exchange Express Edition](image.png)
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 a 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.
### Step 4.3.4

The configuration displayed in the **General Settings** tab for this conference reservation is correlated with the configuration administered for the end user account provisioned in **Step 4.3.2**. Any updates made in this screen will be reflected in the corresponding end user account and vice-versa. To modify parameters associated with this conference reservation, click the **Behavior Definition** tab.

![Reservation Edit](image)

<table>
<thead>
<tr>
<th>General Settings</th>
<th>Behavior Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Conference Name</td>
<td>Reservation for Sample End User Account</td>
</tr>
<tr>
<td>Conference Subject</td>
<td></td>
</tr>
<tr>
<td>Conference Owner Email</td>
<td><a href="mailto:user1@company1.com">user1@company1.com</a></td>
</tr>
<tr>
<td>Type</td>
<td>Demand</td>
</tr>
<tr>
<td>Conference Profile</td>
<td>Full Featured Demand with Recording</td>
</tr>
<tr>
<td>Seats</td>
<td>250</td>
</tr>
<tr>
<td>Language</td>
<td>English</td>
</tr>
<tr>
<td>Passcode Settings</td>
<td></td>
</tr>
<tr>
<td>Auto Generate Passcodes is ON. (To generate new passcodes remove existing passcodes.)</td>
<td></td>
</tr>
<tr>
<td>Moderator Passcode</td>
<td>444</td>
</tr>
<tr>
<td>Participant Passcode</td>
<td>144</td>
</tr>
</tbody>
</table>

---

REB; Reviewed: SPOC 11/19/2007  Solution & Interoperability Test Lab Application Notes  ©2007 Avaya Inc. All Rights Reserved.  S6100AcmlImgCas
The configuration displayed in the **Behavior Definition** tab may be modified to suit the requirements for this conference. For this sample configuration, a blast dial list was provisioned. To configure a blast dial list, click the **Blast Dial List** button.
From the **Blast Dial List** screen, add entries to the blast dial list as follows:

- Enter a number in the **Phone Number** field that is associated with the following:
  - The telephone number pattern provisioned for the TelNum to URI map in **Step 4.2.5**.
  - Telephones registered to either Avaya Communication Manager, or Avaya SIP Enablement Services.
- Enter a descriptive name for this phone number in the **Name** field.
- Click the **Add** button to add entries to this blast dial list.
- The resultant provisioning is shown below.

![Blast Dial List](image.png)
### Step 4.3.7
Repeat **Step 4.3.6** to add additional phone numbers to the blast dial list. The resultant blast dial 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.

### Blast Dial List

<table>
<thead>
<tr>
<th>Phone Number</th>
<th>Name</th>
</tr>
</thead>
<tbody>
<tr>
<td>31002</td>
<td>SIP-31002</td>
</tr>
<tr>
<td>32002</td>
<td>Digital-31002</td>
</tr>
<tr>
<td>33002</td>
<td>H323-31002</td>
</tr>
<tr>
<td>34002</td>
<td>Analog-31002</td>
</tr>
</tbody>
</table>

**Add**

**Blast Dial Users**
5. Cantata Technology IMG 1010 Configuration

This section displays the configuration for enabling the IMG to interoperate with Avaya Communication Manager as well as Avaya Meeting Exchange.

The IMG was administered from the Cantata Technology ClientView (ClientView) application running which was co-resident with the Cantata Technology GateControl Element Management System (GCEMS) running on a Linux server. Refer to the Cantata website for on-line documentation regarding the IMG, GCEMS and the ClientView application.

Note that this section displays the provisioning that was utilized for this sample configuration, and does not show exhaustive procedures for administering an initial configuration. For example, the screens for adding “new” elements to this sample configuration are not shown. However, the sequence of these procedures is relevant, as the configuration was administered in the order presented. Refer to the on-line help available on the Cantata website regarding procedures/commands to administer an initial configuration.
Figure 2 illustrates the main window of the ClientView application that was utilized to provision the IMG. The following panes appear in the main window:

- The **Configuration Tree**, which is located in the top-left portion of the main window. This pane contains all of the items that can be configured. Right-click an item to access additional configuration items. Creating an entry in the Configuration Tree opens the corresponding Configuration Pane.

- The **Configuration Pane**, which is located in the top-right portion of the main window. This pane shows the properties of the selected object. This pane is used to view and edit the configuration.
  - The column titled **As-Configured**, shows the current configuration for parameters, as defined by the **Property** column. Enter or edit values in the **User-Specified** column.

![Figure 2: Cantata Technology ClientView Main Window](image-url)
<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 5.1.1 | A default configuration file named “default” is created when ClientView connects to GCEMS. To save the configuration file with a new name:  
  - Right-click **Configuration default** in the Configuration Tree, and select **Modify**.  
  - Enter a descriptive name in the **Filename** field in the Configuration Pane.  
  - To save the changes, right-click **Configuration DevConnect**, and select **Commit**.  
  - The resultant provisioning is shown below. |

![Configuration screen capture](image_url)
Step 5.1.2: Create a logical IMG as follows:

- Right-click **Cantata IMG EMS** in the Configuration Tree, and select **New Logical IMG**.

  ![Cantata IMG EMS](image)

  ![New Logical IMG](image)

- Enter a descriptive name for the logical IMG in the **Name** field in the Configuration Pane.

- To save the changes, right-click **Logical IMG Avaya-IMG**, and select **Commit**.

- The resultant provisioning is shown below.
5.1.3 Create a physical IMG as follows:

- Right-click the logical IMG in the Configuration Tree, and select **New Physical IMG**.
- Enter a descriptive name for the physical IMG in the **IMG Name** field in the Configuration Pane.
- Enter the IP address of the physical IMG in the **IP Address** field. This is the same IP address assigned to the CTRL 0 port on the back of the IMG.
- Use default settings for remaining fields.
- To save the changes, right-click **IMG Name: Avaya-IMG - ID:0**, and select **Commit**.
- The resultant provisioning is shown below.
5.1.4 Create an object for Network Interfaces as follows:

- Right-click the physical IMG in the Configuration Tree, and select **New Network Interfaces**.
- To save the changes, right-click **Network Interfaces**, and select **Commit**.
- The resultant provisioning is shown below.
Create a Network Interface corresponding to VoIP Module 0: Port 0 as follows:

- Right-click **Network Interfaces** in the Configuration Tree, and select **New Network Interface**.
- Select **VoIP Module 0: Port 0** from the drop down list for the **Physical Interface** field in the Configuration Pane.
- Administer settings for module’s IP network configuration in the **IP Address**, **Subnet** and **Default Gateway** fields respectively.
- Use default settings for remaining fields.
- To save the changes, right-click **VoIP Module 0: Port 0**, and select **Commit**.
- The resultant provisioning is shown below.
Step 5.1.6 Create a Network Interface corresponding to the CPU as follows:

- Right-click **Network Interfaces** in the Configuration Tree, and select **New Network Interface**.
- Select **CPU** from the drop down list for the **Physical Interface** field in the Configuration Pane.
- Administer settings for module’s IP network configuration in the **IP Address**, **Subnet** and **Default Gateway** fields respectively.
- Use default settings for remaining fields.
- To save the changes, right-click **CPU**, and select **Commit**.
- The resultant provisioning is shown below.
Step 5.1.7

Create an object for a Facility as follows:

- Right-click the physical IMG in the Configuration Tree, and select **New Facility**.
- To save the changes, right-click **Facility**, and select **Commit**.
- The resultant provisioning is shown below.
Configure VoIP Facilities as follows:

- Right-click Facility in the Configuration Tree, and select New Bearer - IP.
- Use default settings for all fields.

**Note:** The Network IP Address field is populated from the configuration provided for VoIP Module 0: Port 0 in Step 5.1.5.

- To save the changes, right-click VoIPModule 0, and select Commit.
- The resultant provisioning is shown below.
Configure a TDM DS1 as follows:

- Right-click Facility in the Configuration Tree, and select **New TDM DS1**.
- Select **Bearer** from the drop down list for the **Component ID** field.
- Use default settings for remaining fields.
- To save the changes, right-click **Bearer - ID:1**, and select **Commit**.
- The resultant provisioning is shown below.
Configure a T1 Physical Span for CAS as follows

- Right-click the TDM DS1 created in Step 5.1.9 in the Configuration Tree, and select New T1 Physical Span.
- Select CAS from the drop down list for the Signaling field in the Configuration Pane.
- Administer settings for the Framing and Line Coding fields that correspond to the configuration on Avaya Communication Manager (see Step 3.2.1).
- Use default settings for remaining fields.
- To save the changes, right-click CAS - ESF, and select Commit.
- The resultant provisioning is shown below.
5.1.11 Create an object for Signaling as follows:

- Right-click the physical IMG in the Configuration Tree, and select **New Signaling**.
- To save the changes, right-click **Signaling**, and select **Commit**.
- The resultant provisioning is shown below.
Configure SIP Signaling as follows:

- Right-click **Signaling** in the Configuration Tree, and select **New SIP**.
- Administer settings in the Configuration Pane that enable SIP connectivity between the IMG and other SIP User Agents as follows:
  - Enter the IP address assigned to the IMG in the **SIP Signaling IP Address** field.
- Enter values in the **Local SIP Port** and **Default Transport Type** fields that correspond to the configuration on Avaya Meeting Exchange (see **Step 4.2.5**).
- Use default settings for remaining fields.
- To save the changes, right-click **SIP Signaling**, and select **Commit**.
- The resultant provisioning is shown below.
5.1.13 Configure settings for Media as follows:

- Right-click the physical IMG in the Configuration Tree, and select **New Media**.
- Select the Network File Server (NFS) from the drop down list for the **Media Name** field in the Configuration Pane.
- Enter the User ID of the NFS for UNIX permissions in the **User ID** field.
- Enter the Group ID of the NFS for UNIX permissions in the **Group ID** field.
- Use default settings for remaining fields.

*Note: The **Network Interface** field is automatically populated with the IP address provisioned for the management interface for the IMG.*

- To save the changes, right-click **Media IMG0**, and select **Commit**.
- The resultant provisioning is shown below.

---

### ClientView Version 10.3.3.184, Media IMG0 (USER (admin))

**Media IMG0**

- **Property**
  - As-Configured
  - User-Specified
- **Media Name (Used in NFS)**
  - IMG0
- **User ID**
  - 10001
- **Group ID**
  - 100
- **Network Interface**
  - 09192.168.11.111
- **Primary Server ID**
  - None
- **Secondary Server ID**
  - None
- **Primary NFS Server Status**
  - Primary NFS Server Not Configured
- **Secondary NFS Server Status**
  - Secondary NFS Server Not Configured

---

**Client / Server Monitor**

- **Outgoing Trunks**
- **Incoming Trunks**
- **License verified for uniqueName<Avaya-IMG>**
Step 5.1.14

Create an object for a Media Module as follows:

- Right-click **Media IMGO** in the Configuration Tree, and select **New Media Module**.
- Use default settings for all fields.
- To save the changes, right-click **Media Module 0**, and select **Commit**.
- The resultant provisioning is shown below.
Configure the Media Module DSP as follows:

- Right-click the Media Module created in Step 5.1.14 in the Configuration Tree, and select New Media DSP.
- Use default settings for all fields.
- To save the changes, right-click Media DSP 0, and select Commit.
- The resultant provisioning is shown below.
**Step 5.1.16**

Create an object for Profiles as follows:

- Right-click **Cantata IMG EMS** in the Configuration Tree, and select **New Profiles**.
- To save the changes, right-click **Profiles**, and select **Commit**.
- The resultant provisioning is shown below.

![Diagram showing object creation and configuration](attachment:image.png)
Configure an IP Bearer Profile corresponding to Avaya Meeting Exchange as follows:

- Right-click **Profiles** in the Configuration Tree, and select **New IP Bearer Profile**.
- Enter a descriptive name for the IP Bearer Profile in the **IP Bearer Profile Name** field in the Configuration Pane.
- Use default settings for remaining fields.
- To save the changes, right-click **Bearer: MX - ID:2**, and select **Commit**.
- The resultant provisioning is shown below.
### Step 5.1.18

Assign a codec to the IP Bearer Profile corresponding to Avaya Meeting Exchange as follows:

- Right-click the IP Bearer Profile created in Step 5.1.17 in the Configuration Tree, and select **New Supported Vocoders**.
- Select a codec from the drop down list for the **Payload Type** field that is supported on Avaya Meeting Exchange in the Configuration Pane.
- Use default settings for remaining fields.
- To save the changes, right-click **Profile: 2 - Entry:0**, and select **Commit**.
- The resultant provisioning is shown below.

![Configuration Pane with IP Bearer Profile](image)

<table>
<thead>
<tr>
<th>Property</th>
<th>As-Configured</th>
<th>User-Specified</th>
</tr>
</thead>
<tbody>
<tr>
<td>EntryID</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>Payload Type</td>
<td>G711 uaw</td>
<td>G711 uaw</td>
</tr>
<tr>
<td>Preferred Payload Size (ms)</td>
<td>20</td>
<td>20</td>
</tr>
<tr>
<td>Minimum Payload Size (ms)</td>
<td>10</td>
<td>10</td>
</tr>
<tr>
<td>Default Payload Type</td>
<td>Not Used</td>
<td>Not Used</td>
</tr>
<tr>
<td>Annex B Support</td>
<td>Not Used</td>
<td>Not Used</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>EntryID</th>
<th>Payload Type</th>
<th>Preferred Payload Size (ms)</th>
<th>Minimum Payload Size (ms)</th>
<th>Default Payload Type</th>
<th>Annex B Support</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>G711 uaw</td>
<td>20</td>
<td>10</td>
<td>Not Used</td>
<td>Not Used</td>
</tr>
</tbody>
</table>
5.1.19 Create an object for Signaling Variants as follows:

- Right-click **Cantata IMG EMS** in the Configuration Tree, and select **New Signaling Variants**.
- To save the changes, right-click **Signaling Variants**, and select **Commit**.
- The resultant provisioning is shown below.
Step | Description
--- | ---
5.1.20 |Configure a Signaling Variant to enable CAS connectivity with Avaya Communication Manager as follows:
- Right-click **Signaling Variants** in the Configuration Tree, and select **New Signaling Variant**.
- Enter a descriptive name for the Signaling Variant in the **Variant Name** field in the Configuration Pane.
- Select **CAS** from the drop down list for the **Variant Type** field.
- Use default settings for remaining fields.
- To save the changes, right-click **CM-CAS**, and select **Commit**.
  - Right-click on **CM-CAS** to add objects. For this sample configuration, the objects shown in the configuration tree were added.
- The resultant provisioning is shown below.
Step 5.1.21

Modify the Inpulsing Parameters object as follows:

- Right-click the Inpulsing Parameters object in the Configuration Tree.
- Select **Stage 1** from the drop down list for the **Stage Number** field.
- Use default settings for remaining fields.
- To save the changes, right-click **Inpulsing Parameters Stage 1**, and select **Commit**.
- The resultant provisioning is shown below.

![Configuration Diagram](image-url)
Step | Description
--- | ---
5.1.22 | Create an object for External Network Elements as follows:
- Right-click **Cantata IMG EMS** in the Configuration Tree, and select **New External Network Elements**.
- To save the changes, right-click **External Network Elements**, and select **Commit**.
- The resultant provisioning is shown below.
Create an object for External Gateways as follows:

- Right-click **External Network Elements** in the Configuration Tree, and select **New External Gateways**.
- To save the changes, right-click **External Gateways**, and select **Commit**.
- The resultant provisioning is shown below.
**Step 5.1.24** Configure an External Gateway corresponding to Avaya Meeting Exchange as follows:

- Right-click **External Gateways** in the Configuration Tree, and select **New External Gateway**.
- Enter a descriptive name for the IP Bearer Profile in the **Name** field in the Configuration Pane.
- Select SIP from the drop down list for the **Gateway Signaling Protocol** field.
- Enter the IP address of Avaya Meeting Exchange in the **Gateway IP Address** field.
- Use default settings for remaining fields.

**Note:** The settings for the **Gateway Transport Type**, and **Gateway Remote Port** fields are compatible with the configuration on Avaya Meeting Exchange (see Step 4.1.1, and Step 4.2.5).

- To save the changes, right-click **MX**, and select **Commit**.
- The resultant provisioning is shown below.

---

### MX Configuration Details

<table>
<thead>
<tr>
<th>Property</th>
<th>As-Configured</th>
<th>User-Specified</th>
</tr>
</thead>
<tbody>
<tr>
<td>Name</td>
<td>MX</td>
<td>MX</td>
</tr>
<tr>
<td>Gateway/Signaling Protocol</td>
<td>SIP</td>
<td>SIP</td>
</tr>
<tr>
<td>Gateway/Address Type</td>
<td>Gateway/IP Address</td>
<td>Gateway/IP Address</td>
</tr>
<tr>
<td>Gateway/IP Address</td>
<td>00.192.168.11.121</td>
<td>00.192.168.11.121</td>
</tr>
<tr>
<td>Gateway/Remote Port</td>
<td>5060</td>
<td>5060</td>
</tr>
<tr>
<td>Gateway/Registration Required</td>
<td>No</td>
<td>No</td>
</tr>
<tr>
<td>Registration Expire Interval (sec)</td>
<td>3600</td>
<td>3600</td>
</tr>
<tr>
<td>SIP Profile</td>
<td>Default Profile</td>
<td>Default Profile</td>
</tr>
<tr>
<td>OPTIONS: Keep Alive</td>
<td>Disable</td>
<td>Disable</td>
</tr>
<tr>
<td>Throttles</td>
<td>Yes</td>
<td>Yes</td>
</tr>
<tr>
<td>Privacy</td>
<td>Off</td>
<td>Off</td>
</tr>
</tbody>
</table>

---

### Client View

**Object Name:** MX  
**State:** Configured

- **Configuration**
  - Caller Name: SMC  
  - Port State: Configured
  - Configuration Tree: Click to expand
  - External Gateways: Click to expand

---

[Image of ClientView showing MX configuration details]
### Step 5.1.25

**Description**

Create an object for Routing Configuration as follows:

- Right-click **Cantata IMG EMS** in the Configuration Tree, and select **New Routing Configuration**.
- To save the changes, right-click **Routing Configuration**, and select **Commit**.
- The resultant provisioning is shown below.

![Image of Routing Configuration](image-url)
### Step 5.1.26

Create an object for Channel Groups as follows:

- Right-click **Routing Configuration** in the Configuration Tree, and select **New Channel Groups**.
- To save the changes, right-click **Channel Groups**, and select **Commit**.
- The resultant provisioning is shown below.

---

**Table: Channel Groups**

<table>
<thead>
<tr>
<th>ID</th>
<th>Name</th>
<th>SignalingType</th>
<th>Incoming C</th>
<th>Incoming A</th>
<th>Outgoing C</th>
<th>Outgoing A</th>
<th>Average Hit</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>CAgent</td>
<td>BSN</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>1</td>
<td>CAgent</td>
<td>CAS</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
<tr>
<td>2</td>
<td>MAgent</td>
<td>SP</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
</tbody>
</table>

---

**Diagram:**

- Right-click Routing Configuration in the Configuration Tree, and select New Channel Groups.
- To save the changes, right-click Channel Groups, and select Commit.
- The resultant provisioning is shown below.

---

**Image:**

- Right-click Routing Configuration in the Configuration Tree, and select New Channel Groups.
- To save the changes, right-click Channel Groups, and select Commit.
- The resultant provisioning is shown below.
5.1.27 Configure a Channel Group corresponding to Avaya Communication Manager as follows:

- Right-click Channel Groups in the Configuration Tree, and select New Channel Group.
- Enter a descriptive name for the Channel Group in the Name field in the Configuration Pane.
- Select CAS from the drop down list for the Signaling Type field.
- Use default settings for remaining fields.

**Note:** The administration for the Route Table field is displayed in this screen capture, although the Route Table has not been created. When providing the IMG with an initial configuration, create a Channel Group first, then create a Route Table, then edit the Channel Group to include the Route Table.

- To save the changes, right-click CMeas - ID: 1, and select Commit.
- The resultant provisioning is shown below.
<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 5.1.28 | Create an object for Channel Associated Signaling as follows:  
  - Right-click the Channel Group created in Step 5.1.27 in the Configuration Tree, and select **New Channel Associated Signaling**.  
  - Select the CAS Variant provisioned in Step 5.1.20 from the drop down list for the **CAS Variant** field.  
  - To save the changes, right-click **Channel Associated Signaling**, and select **Commit**.  
  - The resultant provisioning is shown below. |

---

![Diagram of Channel Associated Signaling](ClientView.png)
Assign Channels to the CAS Channel Group corresponding to Avaya Communication Manager as follows:

- Right-click **Channel Associated Signaling** in the Configuration Tree, and select **New CAS Circuits**.
- Select **Bearer** from the drop down list for the **IMG Interface** field.
- Use default settings for remaining fields.
- To save the changes, right-click **CAS Channels: Bearer-1**, and select **Commit**.
- The resultant provisioning is shown below.

**Note:** The IMG counts channels from zero, where Avaya Communication Manager counts from one.
<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 5.1.30 | Configure a Channel Group corresponding to Avaya Meeting Exchange as follows:  
- Right-click **Channel Groups** in the Configuration Tree, and select **New Channel Group**.  
- Enter a descriptive name for the Channel Group in the **Name** field in the Configuration Pane.  
- Select **SIP** from the drop down list for the **Signaling Type** field.  
- Use default settings for remaining fields.  
  *Note: The administration for the **Route Table** field is displayed in this screen capture, although the **Route Table** has not been created. When providing the IMG with an initial configuration, create a **Channel Group** first, then create a **Route Table**, then edit the **Channel Group** to include the **Route Table**.*  
- To save the changes, right-click **MXsip - ID: 3**, and select **Commit**.  
- The resultant provisioning is shown below. |
### Step 5.1.31

**Description**

Assign an IP Network Element to the Channel Group corresponding to Avaya Meeting Exchange as follows:

- Right-click the Channel Group created in **Step 5.1.30** in the Configuration Tree, and select **New IP Network Element**.
- Select the External Gateway provisioned in **Step 5.1.24** from the drop down list for the IP Network Element field.
- To save the changes, right-click **IP Network Element MX**, and select **Commit**.
- The resultant provisioning is shown below.

#### Diagram

![ClientView Version 16.1.1.24 - IP Network Element MX (USER: admin)_ROE - (Monitor / Provision / Configure)](image)

<table>
<thead>
<tr>
<th>Property</th>
<th>As-Configured</th>
<th>User-Specified</th>
</tr>
</thead>
<tbody>
<tr>
<td>IP Network Element</td>
<td>MX</td>
<td>MX</td>
</tr>
</tbody>
</table>

**REB; Reviewed:** SPOC 11/19/2007

__Solution & Interoperability Test Lab Application Notes__

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5.1.32 **Create an object for Routing Tables as follows:**
- Right-click **Routing Configuration** in the Configuration Tree, and select **New Routing Tables**.
- To save the changes, right-click **Routing Tables**, and select **Commit**.
- The resultant provisioning is shown below.

5.1.33 **Configure a Route Table as follows:**
- Right-click **Routing Tables** in the Configuration Tree, and select **New Route Table**.
- Enter a descriptive name for the Route Table in the **Name** field in the Configuration Pane.
- Use default settings for remaining fields.
- To save the changes, right-click the entry, and select **Commit**. See **Step 5.1.34** for resultant provisioning.
**Step 5.1.34** Add route entries to the Route Table provisioned in **Step 5.1.33** as follows:

- To add a route entry corresponding to Avaya Communication Manager, right-click the **Route Table** in the Configuration Tree and select **Add Route Entry**.
  
  - Enter a pattern to match extensions on Avaya Communication Manager, where & is a wildcard, in the **Router String** field in the **New Entry** dialog box.
  
  - Select the Channel Group provisioned in **Step 5.1.27** from the drop down list for the **Outgoing Channel Group** field.
  
  *Note: This is displayed below under the Route Action List column.*
  
  - Click **OK** in the **New Entry** dialog box.

- To add a route entry corresponding to Avaya Meeting Exchange, right-click the **Route Table** in the Configuration Tree and select **Add Route Entry**.
  
  - Enter a pattern to match the provisioning for call flows on Avaya Meeting Exchange, where & is a wildcard, in the **Router String** field in the **New Entry** dialog box.
  
  - Select the Channel Group provisioned in **Step 5.1.30** from the drop down list for the **Outgoing Channel Group** field.
  
  *Note: This is displayed below under the Route Action List column.*
  
  - Click **OK** in the **New Entry** dialog box.

- The resultant provisioning is shown below.
6. Interoperability Compliance Testing

6.1. General Test Approach
The general test approach was to place calls between Avaya Communication Manager and Avaya Meeting Exchange via the IMG utilizing the sample configuration displayed in Figure 1. The main objectives were to verify the following:

- Inbound calling from Avaya Communication Manager to scheduled and demand conferences provisioned on Avaya Meeting Exchange via the Cantata IMG 1010:
  - Direct call flow (without participant-access-code)
  - Basic call flow (with participant-access-code)
- Outbound calling from Avaya Meeting Exchange to stations registered to either Avaya Communication Manager, or Avaya SIP Enablement Services via the Cantata IMG 1010:
  - Blast dial to a pre-provisioned blast dial list
  - Originator dial-out
- Conference features 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:
  - Call hold
  - Attended/unattended call transfer
  - Call forward
  - Three-way conference
- The following transport methods for signaling between Avaya Meeting Exchange and the IMG:
  - TCP
  - UDP
- The following transport methods for signaling/media between Avaya Communication Manager and the IMG:
  - T1 CAS (Robbed-Bit)
- The following codecs:
  - G711MU
- Subjective voice quality for endpoints participating in a conference.
- DTMF transmission via RFC 2833.

6.2. Test Results
All test cases, as defined by the general test approach, passed.
7. 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.

<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| **7.1.1** | Verify CAS connectivity between Avaya Communication Manager and the IMG by retrieving status regarding the trunk group provisioned in Step 3.2.2. From a SAT session:  
  - Issue the command “status trunk <n>”, where <n> is the number of the trunk group to verify.  
  - Verify that all members in the trunk group are **in-service/idle**. |

| **7.1.2** | Validate signaling and media connectivity for inbound calls to Avaya Meeting Exchange from Avaya Communication Manager via the IMG. This is accomplished by verifying that the trunk provisioned in Step 3.2.2 is utilized when a call from a phone registered to either Avaya Communication Manager, or Avaya SIP Enablement Services dials in to a conference provisioned on Avaya Meeting Exchange. From a SAT session:  
  - Issue the command “list trace tac <n>”, where <n> is the TAC defined for the trunk group.  
  - From a station registered to either Avaya Communication Manager, or Avaya SIP Enablement Services, dial 444 to enter the conference provisioned in Section 4.3 as moderator via the direct call flow provisioned in Step 4.2.2.  
  
  **Note:** The trace below shows a station (33006) that dialed (444) and utilized the call routing provisioned in Section 3.3 to route the call to Avaya Meeting Exchange. |

```
lst trace tac 107 | Page 1
LIST TRACE

<table>
<thead>
<tr>
<th>time</th>
<th>data</th>
</tr>
</thead>
<tbody>
<tr>
<td>10:29:35</td>
<td>Calling party station 33006 cid 0x289</td>
</tr>
<tr>
<td>10:29:35</td>
<td>Calling Number &amp; Name 33006 H.323 33006 V</td>
</tr>
<tr>
<td>10:29:35</td>
<td>dial 444 route:AAR</td>
</tr>
<tr>
<td>10:29:36</td>
<td>term trunk-group 7 cid 0x289</td>
</tr>
<tr>
<td>10:29:36</td>
<td>dial 444 route:AAR</td>
</tr>
<tr>
<td>10:29:36</td>
<td>route-pattern 7 preference 1 cid 0x289</td>
</tr>
<tr>
<td>10:29:38</td>
<td>seize trunk-group 7 member 11 cid 0x289</td>
</tr>
<tr>
<td>10:29:38</td>
<td>dial 444 route:AAR</td>
</tr>
<tr>
<td>10:29:38</td>
<td>output pulse done 444</td>
</tr>
<tr>
<td>10:29:38</td>
<td>active trunk-group 7 member 11 cid 0x289</td>
</tr>
</tbody>
</table>
```
<table>
<thead>
<tr>
<th>Step</th>
<th>Description</th>
</tr>
</thead>
</table>
| 7.1.3 | Validate signaling and media connectivity for outbound calls from Avaya Meeting Exchange to Avaya Communication Manager via the IMG. This is accomplished by verifying that the trunk provisioned in Step 3.2.2 is utilized when a call is placed from a participant in conference on Avaya Meeting Exchange to a station registered to either Avaya Communication Manager, or Avaya SIP Enablement Services. From a SAT session:  
  • Issue the command “list trace tac <n>”, where n is the TAC defined for the trunk group.  
  • 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.  
  
*Note:* The trace below shows the call that originated from Avaya Meeting Exchange to a SIP station registered to Avaya SIP Enablement Services. The call utilized the trunk group between Avaya Communication Manager and the IMG. |

```
list trace tac 107

LIST TRACE

time       data
10:31:39   Calling party trunk-group 7 member 10 cid 0x28a
10:31:39   Calling Number & Name NO-CPNumber NO-CPName
10:31:39   tone-receiver 01AXX04 cid 0x28a
10:31:39   active trunk-group 7 member 10 cid 0x28a
10:31:40   dial 31002
10:31:40   term station 31002 cid 0x28a
10:31:42   active station 31002 cid 0x28a
```

| 7.1.4 | 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"`  
    o This command provides a real time, continuous update of port utilization on Avaya Meeting Exchange. |
8. Conclusion
These Application Notes presented a compliance-tested solution comprised of Avaya Communication Manager, Avaya Meeting Exchange Express Edition, and the Cantata Technology IMG 1010 Media Gateway. This solution enables connectivity between Avaya Communication Manager and Avaya Meeting Exchange Express Edition via the Cantata Technology IMG 1010 Media Gateway utilizing standards based SIP and CAS connectivity.

9. Additional References
Avaya references are available at [http://support.avaya.com](http://support.avaya.com).
