

## SIP Trunking using the Optimum Business SIP Trunk Adaptor and the CudaTel 2.6.004 IPPBX

### Goal

The purpose of this configuration guide is to describe the steps needed to configure the CudaTel IP PBX version 2.6.004 for proper operation with Optimum Business SIP Trunking.

### Prerequisites

Please follow the instructions in the Optimum Business SIP Trunk Set-up Guide. The Set-up Guide was left by the Optimum Business technician at installation. If you do not have the Set-up Guide, go to [www.optimumbusiness.com/SIP](http://www.optimumbusiness.com/SIP) to download a copy.

### Important:

The Cablevision network supports only inband DTMF tones. The CudaTel PBX supports only sending out-of-band DTMF tones. In order for the CudaTel PBX to operate correctly with the Cablevision network, the Optimum SIP Trunk Adaptor must be enabled to convert out-of-band DTMF tones sent by the CudaTel PBX to inband DTMF tones. To enable this conversion, log into the Optimum SIP Trunk Adaptor using the login and password specified in the Optimum SIP Trunk Adaptor Set-up Guide. On the **SIP Trunk Configuration** page, you **must** check the **Convert Inband DTMF** checkbox, and click the **Submit** button to update this setting. This is Step 3 of the Optimum Sip Trunk Set-up Guide.

This configuration guide provides the configuration steps for both PBX registration and static or non-registration modes of PBX operation.

## PBX Configuration

To access the PBX configuration GUI point the browser to 192.168.200.200 (unless changed from the default IP previously). The below login page will appear.



Figure-1

The default username and password login is:

Username: admin

Password: admin

A screen similar to the below image will be displayed upon successfully logging into the PBX.

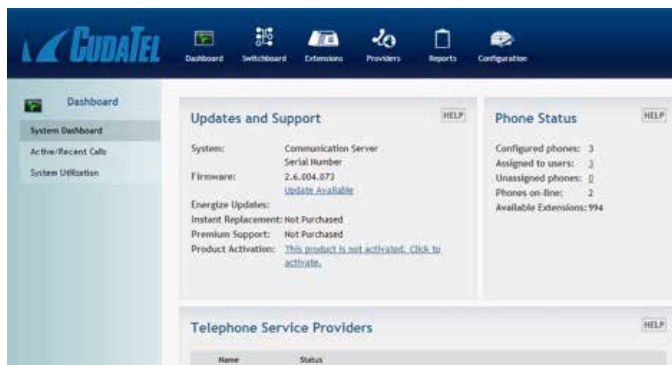


Figure-2

Confirm the firmware is 2.6.004.

Select the **Configuration** button in the upper panel on the far right, select the **Network** link on the left panel.

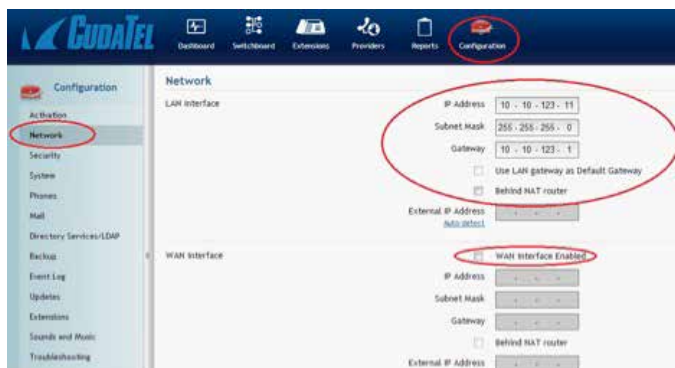


Figure-3

Modify the following settings:

**LAN Interface:** This section will configure the static IP address of the PBX.

**IP Address:** Enter in the IP address to be assigned to the PBX.

**Subnet Mask:** Enter in the subnet mask for the network assigned to the PBX.

**Gateway:** Enter in the Gateway IP address to the network. By default, this is the LAN side IP address of the upstream Optimum Business SIP Trunk Adaptor.

**Use LAN Gateway as Default Gateway:** Uncheck this box.

**Behind NAT Router:** Uncheck this box.

**WAN Interface Enabled:** Uncheck this box. No other fields should be populated.

Scroll to the bottom of the page, click on the **Apply Changes** button.

With this Network configuration, the WAN port on the PBX will no longer be in use. The LAN port will be plugged into the Optimum Business SIP Trunk Adaptor's LAN port or the switch that plugs into the Optimum Business SIP Trunk Adaptor's LAN port.

On the left panel, click the **Phones** link. The page below will appear.

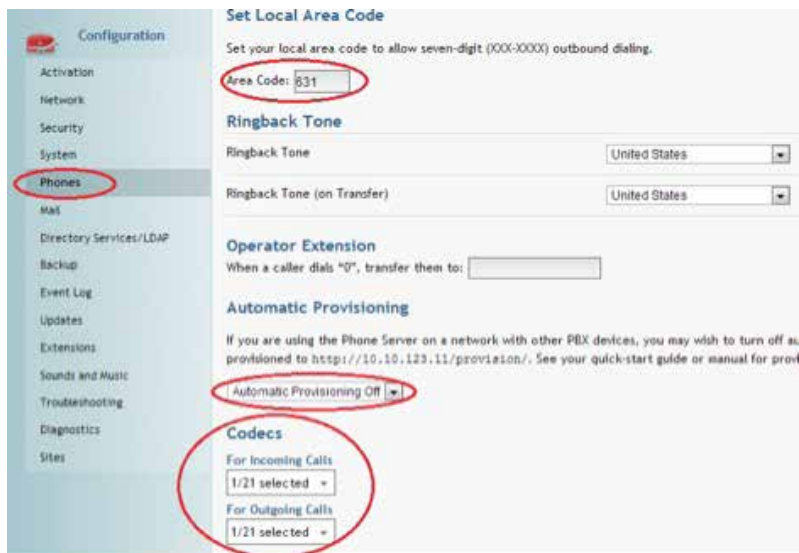


Figure-4

**Set Local Area Code:** Enter in the Area Code the DID's will use.

**Automatic Provisioning:** Set this drop-down box to Automatic Provisioning Off.

**Codecs:** For Incoming Calls, and For Outgoing Calls unselect ALL codecs except the G.711u and G.711a option.

All other options and fields should be left blank or left to the default configuration. On this page there is no Apply Settings options, this is done automatically as soon as the options are set.

Click the **Extensions** link on the left panel. The below page will appear.



Figure-5

Click the **Add an Extension Block** link.

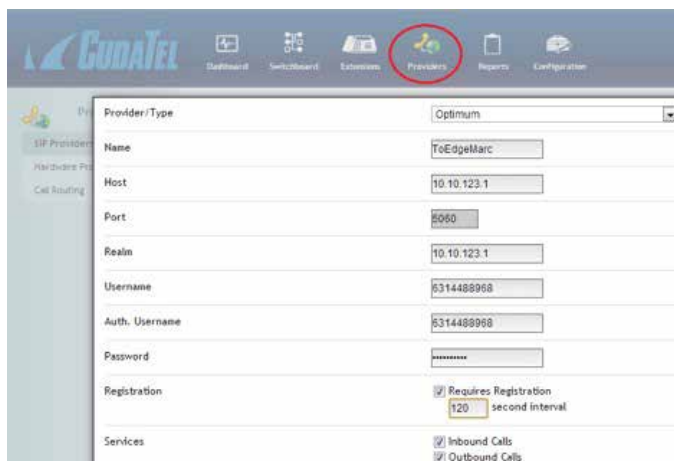
A new field will prompt for an extension range, in the first box enter the starting extension such as 2000, enter 2999 in the second box.

Click the blue **Add New Extension Block** button to complete this change.

On this page there is no Apply Settings button, settings are saved by default.

On the upper panel, click the **Providers** link. Click the **New SIP Account** link. The below window will appear.

**For Registration Mode:** Follow the below settings.



The screenshot shows the CudaTel web interface with the 'Providers' tab selected. The configuration form is as follows:

Field	Value
Provider/Type	Optimum
Name	ToEdgeMarc
Host	10.10.123.1
Port	5060
Realm	10.10.123.1
Username	6314489968
Auth. Username	6314489968
Password	XXXXXXXXXX
Registration	<input checked="" type="checkbox"/> Requires Registration 120 second interval
Services	<input checked="" type="checkbox"/> Inbound Calls <input checked="" type="checkbox"/> Outbound Calls

Figure-6

**Provider/Type:** Select the **Optimum** option from the drop-down list.

**Name:** Set a Name. ToEdgeMarc was selected for this example

**Host:** The LAN IP address of the Optimum Business SIP Trunk Adaptor.

**Port:** Cannot change this setting, defaults to 5060.

**Realm:** Set this to the Optimum Business SIP Trunk Adaptor's LAN IP address, should match the Host field.

**Username:** Pilot DID or assigned username for registration to the Optimum Sip Trunk Adaptor. This must match what was entered in the Optimum Business SIP Trunk Adaptor in Step 3 of the SIP Trunk Set-up Guide.

**Auth. Username:** Same as Username field. This must match what was entered in the Optimum Business SIP Trunk Adaptor in Step 3 of the SIP Trunk Set-up Guide.

**Password:** Registration password. This must match what was entered in the Optimum Business SIP Trunk Adaptor in Step 3 of the SIP Trunk Set-up Guide.

**Registration:** Check the **Require Registration** box and set the second interval field to the recommended value of 120.

**Services:** Check both **Inbound Calls** and **Outbound Calls**.

Leave all other settings as default. Scroll to bottom and click the **Create Gateway** button.

**For Non-Registration Mode:** Follow the below settings.

The screenshot shows the 'New SIP Provider Account' form in the CudaTel interface. The form is titled 'New SIP Provider Account' and has a sidebar with links: SIP Provider, Hardware Pro, and Call Routing. The main form fields are:

- Provider/Type:** Optimum (selected from a drop-down menu)
- Name:** ToEdgeMarc
- Host:** 10.10.123.1
- Port:** 5060
- Realm:** 10.10.123.1
- Username:** (empty field)
- Auth. Username:** (empty field)
- Password:** (empty field)
- Registration:** ☐ Requires Registration (second interval)
- Services:** ☒ Inbound Calls, ☒ Outbound Calls

Figure-7

**Provider/Type:** Select the **Optimum** option from the drop-down menu.

**Name:** Set a Name. ToEdgeMarc was selected for this example.

**Host:** The LAN IP address assigned to the Optimum Business SIP Trunk Adaptor.

**Port:** Cannot change this setting, defaults to 5060.

**Realm:** Set this to the Optimum Business SIP Trunk Adaptor's LAN IP address, should match the Host field.

**Username:** Leave blank.

**Auth. Username:** Leave blank.

**Password:** Leave blank.

**Registration:** Uncheck and Leave blank.

**Services:** Check both **Inbound Calls** and **Outbound Calls**.

Leave all other settings as default. Scroll to bottom and click the **Create Gateway** button.

A new Gateway will appear in the Provider’s page. Click on the new Gateway. The below page will appear.

[BACK](#) **EM-4552** [Rename](#) [Delete](#)

Provider/Type	Optimum
Host	<input type="text" value="10.10.123.1"/>
Port	<input type="text" value="5060"/>
Realm	<input type="text" value="10.10.123.1"/>
Username	<input type="text" value="6314488968"/>
Auth. Username	<input type="text" value="6314488968"/>
Password	<input type="password" value=""/>
Registration	<div><input checked="" type="checkbox"/> Requires Registration 60 second interval <a href="#">Refresh Registration</a> Registered</div>
Services	<div><input checked="" type="checkbox"/> Inbound Calls <input checked="" type="checkbox"/> Outbound Calls <input type="checkbox"/> Faxes</div>

Figure-8



Scroll to the bottom to continue the configuration.

The screenshot displays a configuration page for CudaTel. It includes several sections:
 

- Caller ID Number:** A text field containing '6314408968' and three radio buttons: 'Never use a custom Caller ID number', 'Use a custom Caller ID number unless overridden' (which is selected), and 'Always use a custom Caller ID number'.
- Outgoing Music on Hold:** A dropdown menu set to 'default'.
- Restrict Codecs To:** A dropdown menu set to '2/21 selected'.
- Inbound Registration:** A checkbox labeled 'Allow Inbound Registration' which is currently unchecked.
- Buttons:** An 'Apply Gateway Settings' button is located below the Inbound Registration section.
- External Numbers:** A section showing four numbers: '(631) 448-8968', '(631) 448-8969', '(631) 448-8970', and '(631) 448-8971'. Below them is a link 'Add External Numbers'.
- Outbound Routing:** A section with four radio buttons: '10 Digit Dialing' (selected), '7-digit Dialing (Area Code 631)', 'International Dialing (911)', and 'Emergency (USA)'. Below this is a link 'Manage Routes'.

Figure-9

**Caller ID Number:** In this field enter the Pilot DID. Select the **Use a Custom Caller ID Number Unless Overridden** radio button. This setting will use the Pilot DID by default unless an extension specifically overrides it. To force the extensions Caller ID, select the **Never Use a Custom Caller ID Number**. To always force the Pilot DID on outbound calls select the Always **Use a Custom Caller ID Number**.

**Outgoing Music on Hold:** There are two options, default and silence. Set this option to determine what the remote user hears when put on hold.

**Restrict Codecs To:** Select G.711 ulaw and G.711 alaw.

**Inbound Registration:** Uncheck for security reasons.

**External Numbers:** Click the **Add External Numbers** link. Starting with the Pilot DID, add each DID the PBX will use.

**Outbound Routing:** Leave to default, this will auto-populate.

Click the **Apply Gateway Settings** button.

In the upper panel, click the **Extensions** link. In the new left panel, click the **People** link then click the **Add New Person** link. The below window will appear.

Figure-10

Fill in the **First Name**, **Last Name**, and a **4 digit PIN** for the new extension. Leave all other options to default. Click the blue **Add** button.

After the creation of the user, click on the new user listed under the **People** link and the below page will appear.

Figure-11

**Contact Information (Optional):** Click the **Edit** link to modify the user's contact details.

**Phones:** Click the **Add a Phone** button, the below will appear.

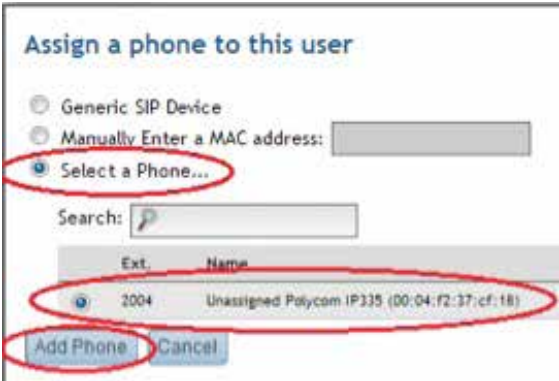


Figure-12

Click the **Select a Phone** button, click an unassigned phone listed. Click the blue **Add Phone** button to complete. The User page should reappear with the phone listed.

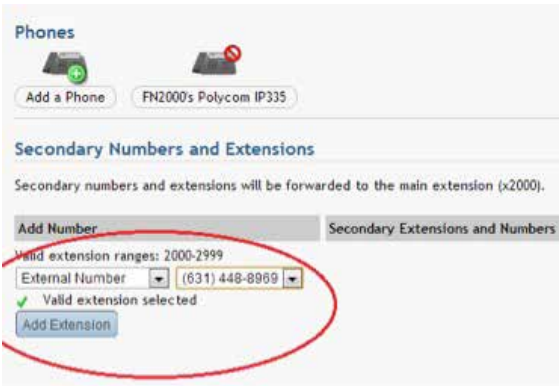


Figure-13

Under the **Add Number** section select **External Number** in the drop-down box. A second drop-down box will appear, select a DID to assign to this extension. Click the **Add Extension** button.

Click the **Apply Setting** button.

Click the **Call Parking Extensions** on the left panel. Then click the **Add New Parking Extension** link.



Figure-14

The below window will appear.

The screenshot shows the 'Add New Parking Extensions' form. The form has the following fields and options:

- Parking Lot Name:** A text input field containing 'Parking'.
- Extension Block:** A text input field containing '700' and a range indicator '- 705'. Below it, a green checkmark and the text 'Valid extension selected' are displayed. A note below the field states: 'Parking extension blocks must be a block of five or more on one of the block's inner extension numbers. The line automatically assigned by transferring to the first exte'.
- Music on Hold:** A dropdown menu with 'default' selected.
- Maximum Hold Time:** A text input field containing '2' and a dropdown menu with 'Minutes' selected. A note below the field states: 'Leave this value blank to enable the feature.'
- After Hold Time, Transfer To...** A text input field containing '2999'. A note below the field states: 'Enter an extension or phone number. Leave this value blank if the Maximum Hold Time above is disabled (blank), (
- Buttons:** 'Add' and 'Cancel' buttons at the bottom. The 'Add' button is circled in red.

Figure-15

**Parking Lot Name:** Name the Parking lot.

**Extension Block:** Enter 700 in the first field, 705 in the second field.

**Music on Hold:** Set to **Default** or **Silence**. The party transferred into the Parking Lot will hear what is defined.

**Maximum Hold Time:** The duration the party is allowed to be parked.

**After Hold Time, Transfer To...:** Where to send the caller after the hold time is up. 2999 in this example is the Automated Attendant (recommended).

Click the blue **Add** button to apply changes.

In the left panel click the **Automated Attendants** then click the **Add New Automated Attendant** link.



Figure-16

The below window will appear.

Figure-17

**Automated Attendant Name:** Give the AA a name.

**Extension:** In the drop-down box select **Single Extension**, enter in the Automated Attendants extension in the field.

Leave other options as default.

Click the blue **Add** button to save the changes.

Click on the new **Automated Attendant** that now appears under the Automated Attendant page. Configure to each option as desired.

**Important:** The DTMF tone duration generated by the phones needs to be increased from the default value of 180ms-200ms to 600ms. The PBX does not have the capability to change the DTMF settings, the change must be done on the phones. The Optimum Business Sip Trunk Adaptor needs to be configured to Convert Inband DTMF. This is Step 3 in the Optimum Business Sip Trunk Set-up Guide.