

MAX Communication Server  
Release 7.5

TelePacific SIP Trunk  
Configuration Guide

April 14, 2015

**Intended audience:**  
**AltiGen Authorized Partners**

## About This Guide

This guide is provided for AltiGen Partners who are provisioning TelePacific SIP Trunks for customers for MaxCS 7.5.

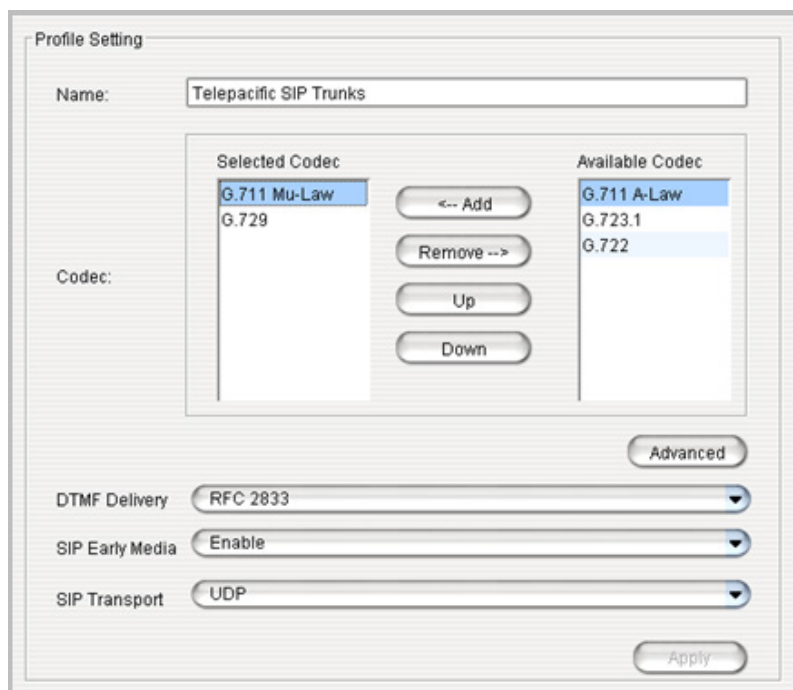
## Requirements

- You must be running MaxCS Release 7.5.

## Create a New Codec Profile

Configure a Codec Profile for the TelePacific SIP trunks in Enterprise Manager.

- In MAXCS Administrator, select **VoIP > Enterprise Network Manager**.
- Click the **Codec** button on the *Quick Launch* bar (this is different from the *IP Codec* tab).
- Create a new profile; click **Add** and create a profile with the following parameters:
  - For the name, enter **Telepacific SIP Trunks**
  - Set the *Selected Codec* to use **G.711 Mu-Law** first and **G.729** second (use the **Add**, **Remove**, **Up**, and **Down** buttons as needed)
  - Set *DTMF Delivery* to **RFC 2833**
  - Set *SIP Early Media* to **Enable**
  - Set *SIP Transport* to **UDP**



Profile Setting

Name:

Codec:

Selected Codec	Available Codec
G.711 Mu-Law	G.711 A-Law
G.729	G.723.1
	G.722

Buttons: <-- Add, Remove -->, Up, Down

Advanced

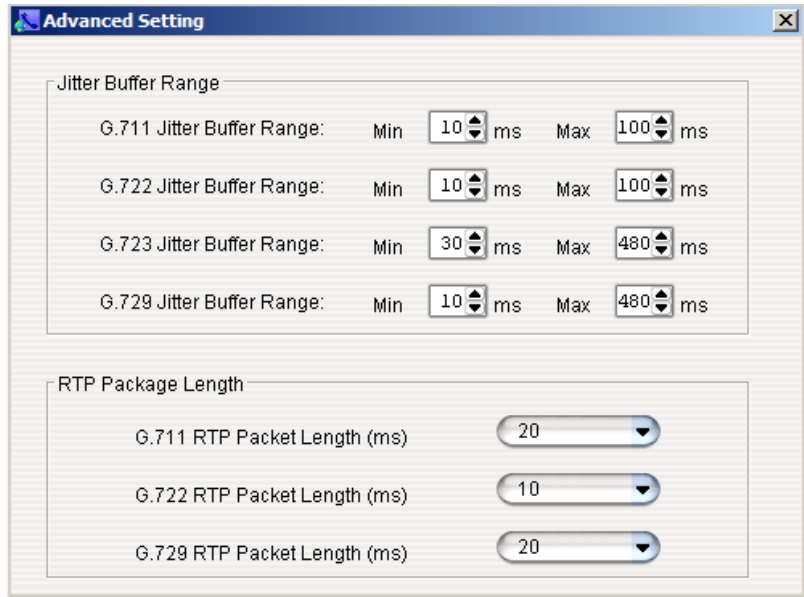
DTMF Delivery:

SIP Early Media:

SIP Transport:

Apply

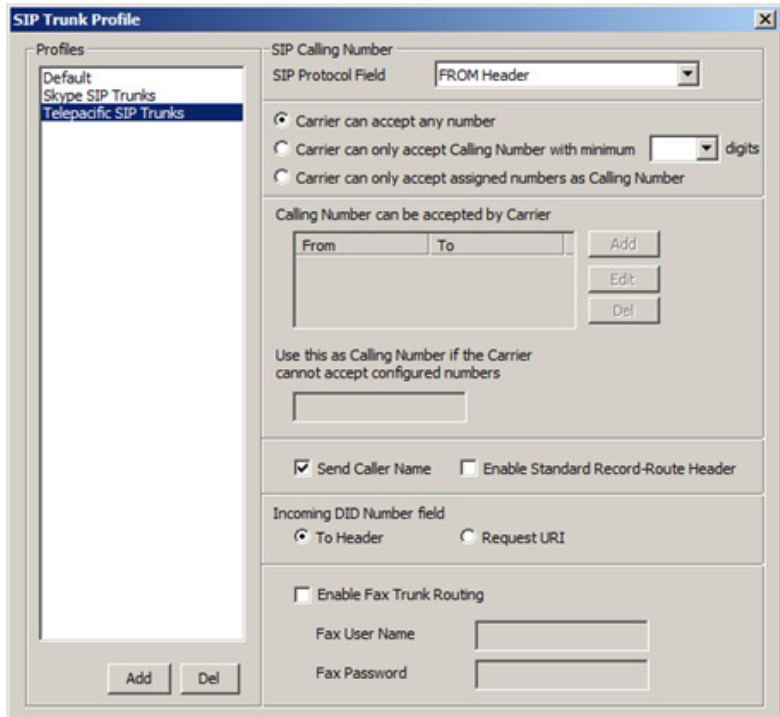
- Click **Advanced** and make sure that your settings match those in the next figure.



## Create a new SIP Trunk Profile

Next, create a new SIP Trunk Profile.

1. In MaxCS Administrator select **Boards**. Double-click **SIPSP**, select **SIP Trunk > SIP Trunk Configuration > SIP Trunk Profile**.
2. Click **Add**. Name the new profile **Telepacific SIP Trunks**.

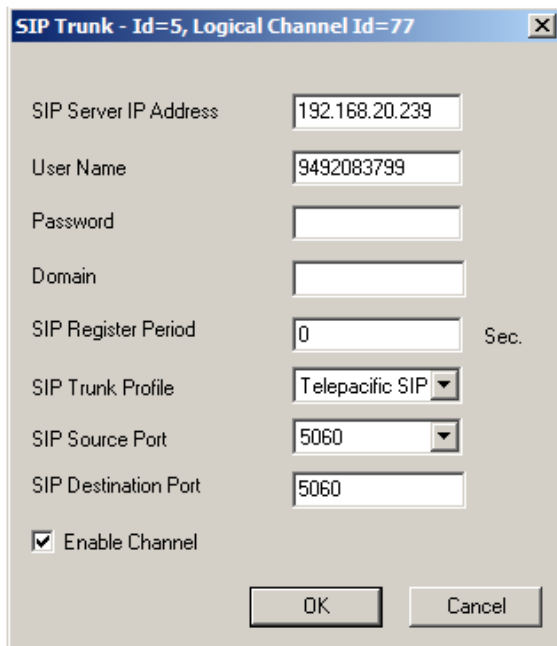


3. Modify the new SIP Trunk profile, specifying the following parameters, and click **OK**.
  - Set *SIP Protocol Field* to **FROM Header**
  - Select *Carrier can accept any number*
  - Select **Send Caller Name**
  - Set *Incoming DID Number* to **To Header**
  - Make sure that the *Enable Fax Trunk Routing* checkbox is not checked

## Configure Channels

Configure channels to use the TelePacific SIP Trunk.

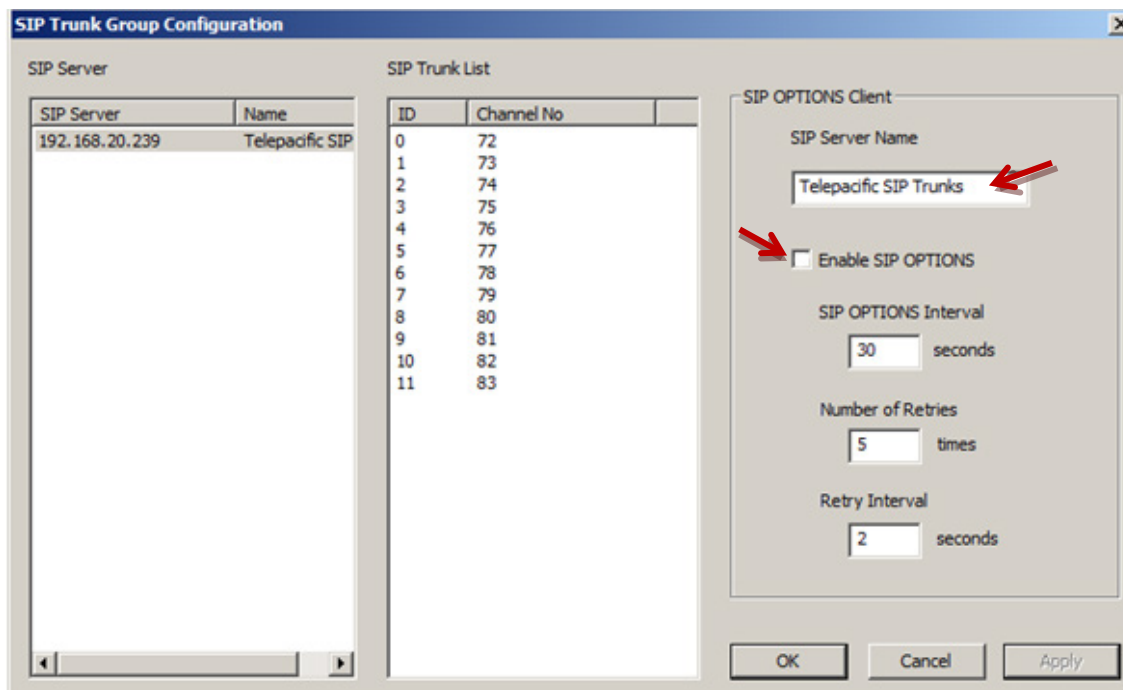
1. Select **Boards**. Double-click **SIPSP**, select **SIP Trunk > SIP Trunk Configuration**.
2. Select the first available SIP Trunk channel and click **Edit**.
3. Modify the parameters as follows, and click **Ok**.
  - For the *SIP Server IP address*, enter the IP address provided by TelePacific; this IP address is their SIP Trunk router
  - For the *User Name*, enter the 10-digit billing telephone number of the SIP Trunk service
  - Leave the *Password* and *Domain* fields blank
  - Set *SIP Registration Period* to **0** seconds
  - For the *SIP Trunk Profile*, select **Telepacific SIP Trunks**
  - Set the *SIP Source Port* and *SIP Destination Port* to **5060**
  - Check the **Enable Channel** checkbox



SIP Server IP Address	192.168.20.239
User Name	9492083799
Password	
Domain	
SIP Register Period	0 Sec.
SIP Trunk Profile	Telepacific SIP
SIP Source Port	5060
SIP Destination Port	5060
<input checked="" type="checkbox"/> Enable Channel	

4. Select the first available SIP Trunk channel and click **Copy To...**

5. Select all SIP channels that will be used for TelePacific SIP Trunks, and click **Ok**.
6. Click **Trunk Group Configuration**. Select the SIP server with the IP address that you entered in step 3 in this section.
  - o For *SIP Server Name*, enter **Telepacific SIP Trunks**
  - o Clear the *Enable SIP OPTIONS* check box
7. Click **Ok**.



## Configure Inbound Routing

Carriers send 10 digits as DNIS; configure your inbound routing rules accordingly.

## Test the SIP Trunks

We recommend that you perform basic tests to confirm that the SIP Trunks are correctly configured; test inbound calls, outbound calls, and extension-to-extension calls.