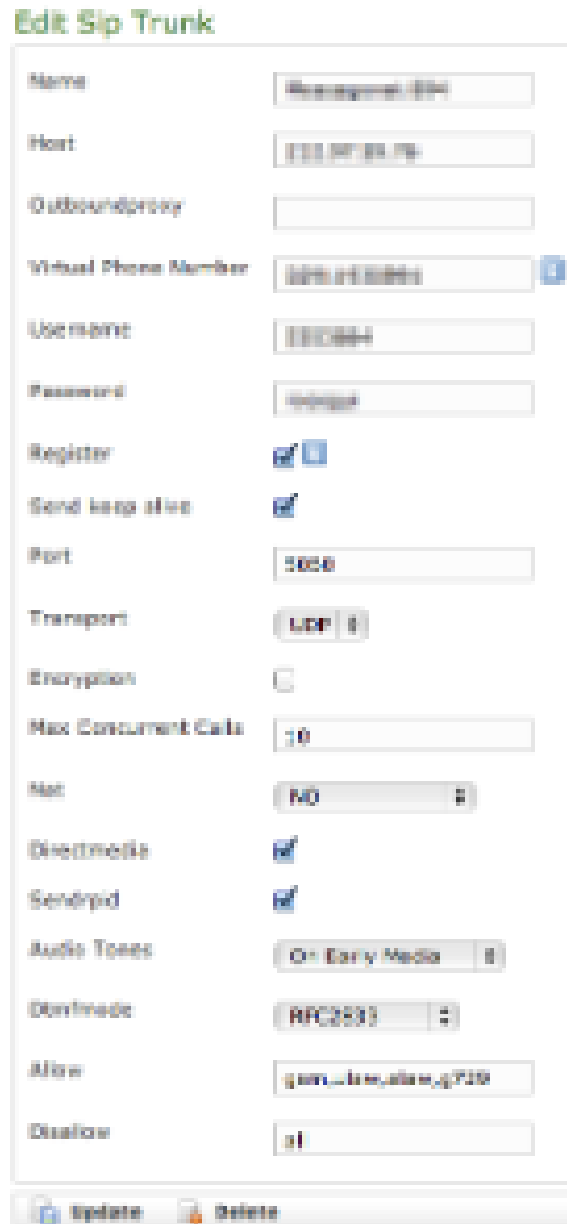


PSAM 4.2 UNENCRYPTED SIP Trunks

Click on the **New Sip Trunk** to go to the Edit Sip Trunk page.



The screenshot shows a web form titled "Edit Sip Trunk" with the following fields and values:

Field	Value
Name	Management SIP
Host	133.207.25.79
Outbound proxy	
Virtual Phone Number	2025.415.0000
Username	2025415
Password	123456
Register	<input checked="" type="checkbox"/>
Send keep alive	<input checked="" type="checkbox"/>
Port	5060
Transport	UDP
Encryption	<input type="checkbox"/>
Max Concurrent Calls	10
Rel	NO
Directmedia	<input checked="" type="checkbox"/>
Sendrpid	<input checked="" type="checkbox"/>
Audio Tones	On Early Media
DTMFmode	RFC2833
Allow	gsm,ulaw,alaw,g719
Disallow	all

At the bottom of the form are two buttons: "Update" and "Delete".

figure 1. "Edit Sip Trunk" form

In figure 1. "Edit Sip Trunk" form you can read an example configuration for an unencrypted SIP trunk. Mandatory fields for the unencrypted configuration are:

- **NAME:** a meaningful name for this trunk
- **HOST:** IP address/hostname of the SIP server provided by ITSP
- **PORT:** this is **5060** by RFC
- **TRANSPORT:** UDP
- **ENCRYPTION:** disabled (unchecked)

We do also suggest the following values to be set:

- **AUDIO TONES: ON EARLY MEDIA** works fine with the Cisco Unified Communications Manager.
- **DTMFMODE:** choose your values considering the PBX on the other end of the Trunk. Usually we suggest to choose the value **RFC2833**
- **DIRECTMEDIA:** enabled (checked)
- **SENDERPID:** enabled (checked)

Other fields in the form depend by your network topology and by the features on the other end PBX.

When you are done with your changes, commit them by clicking on the **Update** icon.

[PSAM 4.1 ENCRYPTED SIP Trunks](#)

[PSAM 4.3 Authentication in SIP Trunks](#)