

How to install our Sip Trunk Product in to your Asterisk server using Username and Password.

You need to add a scrip line in your: etc/asterisk/sip_general_custom.conf

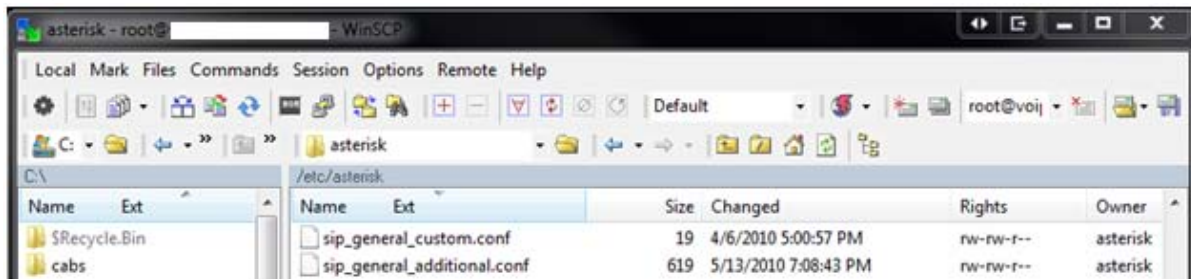
A line: `useragent=DigestPBX`

And save.

What do you need?

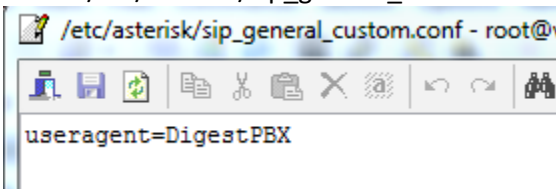
1. You need to have the Root Password.
2. WinSCP software, or any other type of an open source SFTP and FTP client for Microsoft Windows.

Steps:



Login as root using the WinSCP (SFTP and FTP client software) to your Asterisk server.

Go to: /etc/asterisk/sip_general_custom.conf



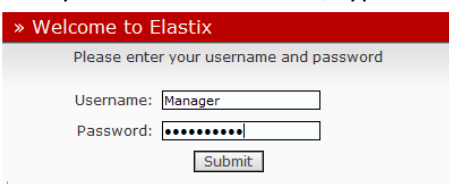
And type: `useragent=DigestPBX`

Press: save  **Press:** Reload 

Exit the WinSCP and go to your Asterisk Web Portal

This example is using Elastix:

Go to your Internet browser, type the IP address of your server.



Login as: Any user with admin privileges.



Press **Trunk** and add a Sip Trunk. [+ Add SIP Trunk](#)

Outbound Caller ID: Account Number

Trunk Name: UsVox

PEER Details:

disallow=all
type=friend
username= **Account number**
secret= **Password**
context=from-trunk
insecure=port,invite
canreinvite=yes
host=sip.usvox.com
allow=g729a&ulaw&gsm
dtmfmode=auto

Register String:

Account number: Password @sip.usvox.com/ Account number

The PEER Details Are common and can be modified as the Administrator please.

If you have experiences working with asterisk just add the lines using the FTP client software.

In the: /etc/asterisk/sip_additional.conf

The changes will look as:

```
[UsVox]
disallow=all
type=friend
username= Account Number
secret= Password
context=from-trunk
insecure=port,invite
canreinvite=yes
host=sip.usvox.com
allow=g729a
allow=ulaw
allow=gsm
dtmfmode=auto
```