



Configure your Asterisk with our telephony IP service

Enjoy the best telephony IP service with Netelip.

Step 1

Register in <http://www.netelip.com/>.

When you're registered, you will receive an e-mail with your SIP account details (user name, password and our voice server domain).

Step 2

Edit file **"sip.conf"** in your Asterisk and add the following lines at the bottom:

```
register => usuario_sip:clave_sip@sip.netelip.com/usuario_sip
```

```
[netelip]
type=friend
username=usuario_sip
context=default
host=sip.netelip.com
canreinvite=no
secret=clave_sip
nat = yes
fromdomain=sip.netelip.com
disallow=all
allow=alaw
insecure=very
fromuser=usuario_sip
```

Step 3

To make calls, add the followings lines in the file **“extensiones.conf”** in the context assigned for switch extensions.

;Salida de llamadas Nacionales

```
exten => _[689]XXXXXXXX,1,Dial (SIP/${EXTEN}@netelip,60)
exten => _[689]XXXXXXXX,n,Hangup()
```

; Salida de llamadas internacionales

```
exten => _00.,1,Dial (SIP/${EXTEN}@netelip,60)
exten => _00.,n,Hangup()
```

To receive calls, add the following lines in the file **“extensiones.conf”** in the context **“default”**:

```
exten => usuario_sip,1,Dial (SIP/extension_interna,60)
exten => usuario_sip,n,Hangup()
```

The concept extension_interna refers to the number or extension name whrere you wish to receive incoming calls.