DoorVaani.com Asterisk Configuration

Tested on Asterisk Version 13.5.0

To configure Asterisk with a DoorVaani.com VOIP Line as a SIP Trunk for making outgoing calls, a couple of files in the Asterisk configuration directory (/etc/asterisk by default) are to be changed as shown below.

File 1: sip.conf

    i.   In the [general] section please add a register statement of the following format using your DoorVaani.com VOIP Line Username and VOIP Line password

register => voip_line_username:voip_line_password@as1.doorvaani.com:5060/voip_line_username

    ii.   Add a new section at the bottom of the file as follows using your own DoorVaani.com VOIP Line username and VOIP Line password

```
[doorvaani]

type=friend
username=voip_line_username
secret=voip_line_password
context=dv-outgoing
host=as1.doorvaani.com
disallow=all
allow=ulaw
allow=alaw
allow=g729
```

File 2: extensions.conf

In your dialplan locate the line containting the call to the Dial function and ensure that the first parameter of the Dial function is like this: SIP/${EXTEN}@doorvaani (see below).

exten => _1XXXXXXXXXX,n,Dial(SIP/${EXTEN}@doorvaani, 30)

Notes:

1. You must reload both sip and dialplan to make the above changes available. Run the following commonds to reload SIP and dialplan at the asterisk command line

   sip reload

   dialplan reload

2. Run the following command as the asterisk command line to see your line registered

   sip show registry