

# Create your first VoiceXML application

To configure your first VoiceXML application inside Asterisk, just edit your dialplan (extensions.conf):

```
# cd /etc/asterisk
# vi extensions.conf
```

Add a new extension in your dialplan (/etc/asterisk/extensions.conf ) like:

```
exten => 888,1,Answer
exten => 888,2,Wait(3)
exten => 888,3,Vxml(http://vxml.i6net.com/index.vxml)
exten => 888,4,Hangup

exten => 999,1,Answer
exten => 999,2,Wait(3)
exten => 999,3,Vxml(file:///tmp/helloworld2.vxml)
exten => 999,4,Hangup
```

Install first demo VoiceXML examples programs in your /tmp directory:

```
helloworld2.vxml ; 1st program with WAV file access says "Hello
World!"
```

Please ensure you have reloaded your extensions in Asterisk using the prompt command:

```
*CLI> extensions reload
```

Call the service by calling :

```
SIP:888@<your server IP address>
SIP:999@<your server IP address>
```

The best way to manage VoiceXML applications is to use the VoiceXML account configuration.

You can catch the errors generated from the Asterisk Vxml application (error is in the variable VXML\_ERROR) to propose an alternative call processing from the Asterisk dialplan.

Example :

```
exten => _X.,n,Vxml()
exten => _X.,n,NoOp(${VXML_ERROR})
exten => _X.,n,GotoIf("${VXML_ERROR}" == "")?hangup)
exten => _X.,n,Busy()
exten => _X.,n(hangup),Hangup()
```

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