

# Voximal Web Configuration

## Access

All the configuration is managed through a web interface based on the FreePBX project. To access to the web interface, open the link

[http://your\\_server\\_address/admin](http://your_server_address/admin).

It's protected by a login (the first login after the installation will request you a password).



## Login

After installation there are 3 default login accounts, with different access permissions

- **user** : access to Voximal configuration and reports.
- **admin** : user access and network configuration
- **root** : full access (full freePBX features).

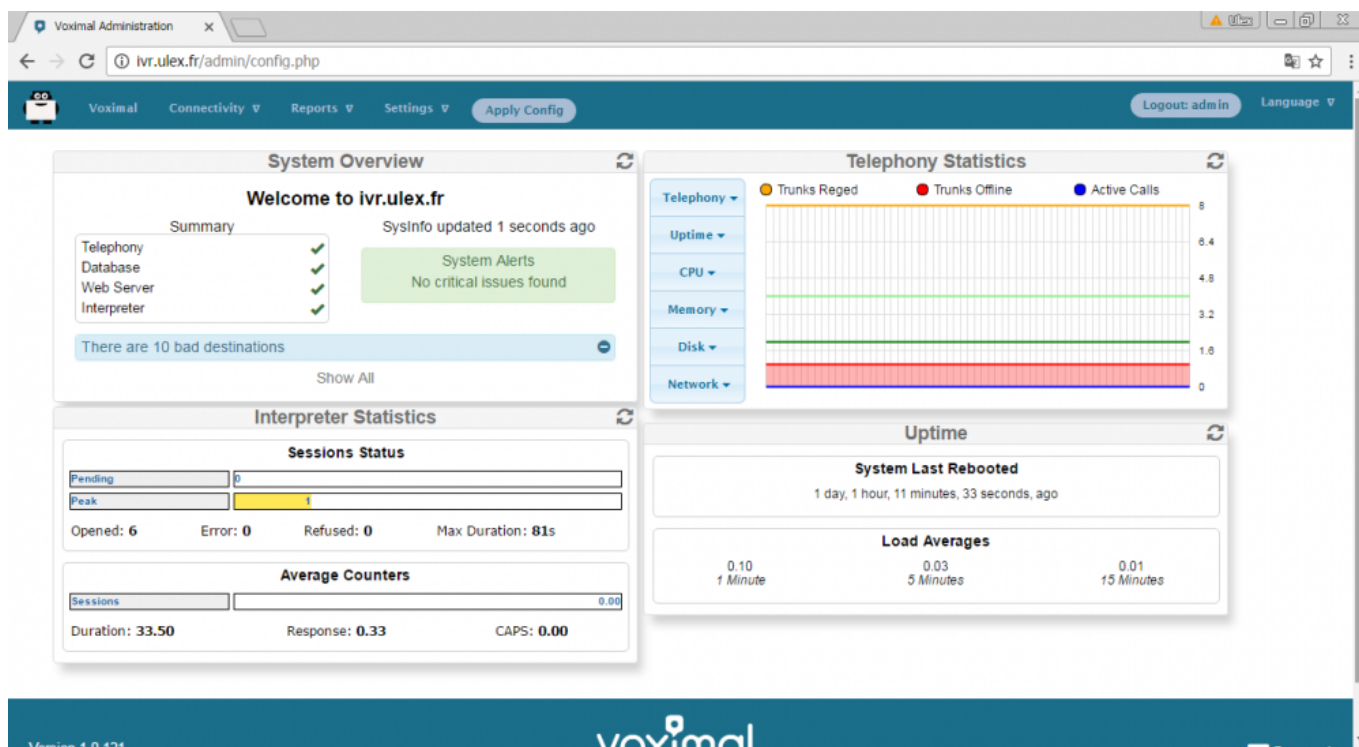
Each login has a default password (requested after the package installation, or the image activation), you can change them with the root account.



## System Status

The home page after login show you the system status.

- Host name of the server
- Summary : Main modules status
- Interpreter statistics
- Telephony statistics
- Uptime / Load average



## Main configuration steps

To configure the Voximal IVR you have 4 steps to do :

1. Configure the VoIP telephony access ([SIP accounts](#))
2. Create/edit a VoiceXML service ([Create a VoiceXML service](#))
3. Create Voximal accounts ([Voximal accounts](#))
4. Configure the relation between the DID and the VoiceXML service you want to use ([Routes configuration](#))

**Note:** On each configuration page, don't forget to click on **[Submit]** button at the bottom page before changing page. If you don't click on **[Submit]** you'll lose all your changes.

**Note:** All changes are not directly configure on the server. After one change, you'll see a button **[Apply Config]** in menu. To activate the changes on server you have to click on **[Apply Config]**.

### 1) Configure the telephony access

#### a) Use the test number and the PIN

If your server is connected to the internet you can use the free test access to place calls to your server. There is nothing to do. You should only check that the port 4569 is open from/to internet in UDP.

## b) Connect a trunk SIP

You can create a SIP trunk with an operator with the page [Voip Providers](#). You have to enter informations :

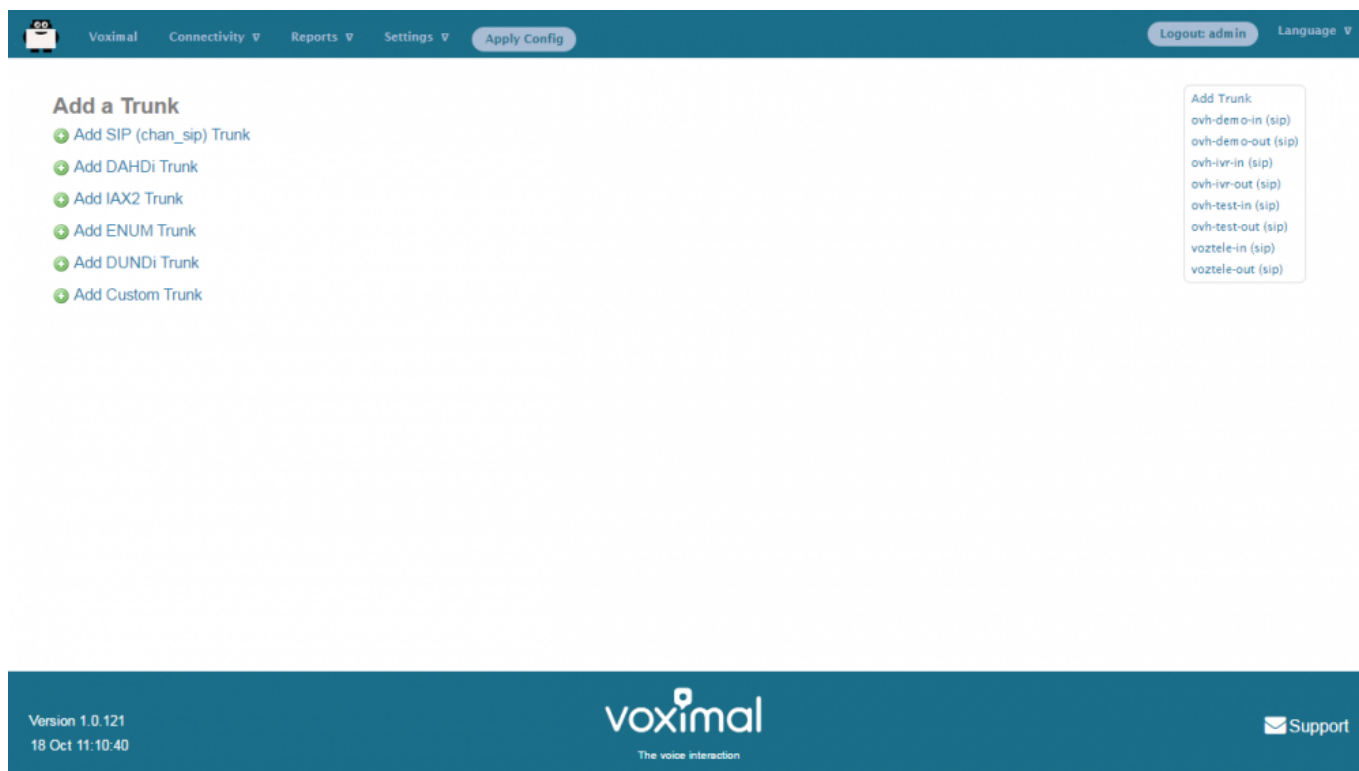
- A trunk name : a string to identify you accounts
- The peers details : enter all trunk informations (host, username, password, type)

```
host=myprovider.com
username=0033123456789
type=peer
```

- Register informations : enter registry informations like :

```
0033123456789:password@myprovider.com/33123456789
```

- Click on **[submit]** button (don't remember to apply your modifications)



## c) Add an extension

After you setup your Voxibot, the first thing you do is to add extensions (connect a Phone). The integrated FreePbx allows you to add a couple of different Device types

- Generic SIP Device
- Generic IAX2 Device
- Generic DAHDi Device
- Other Custom Device

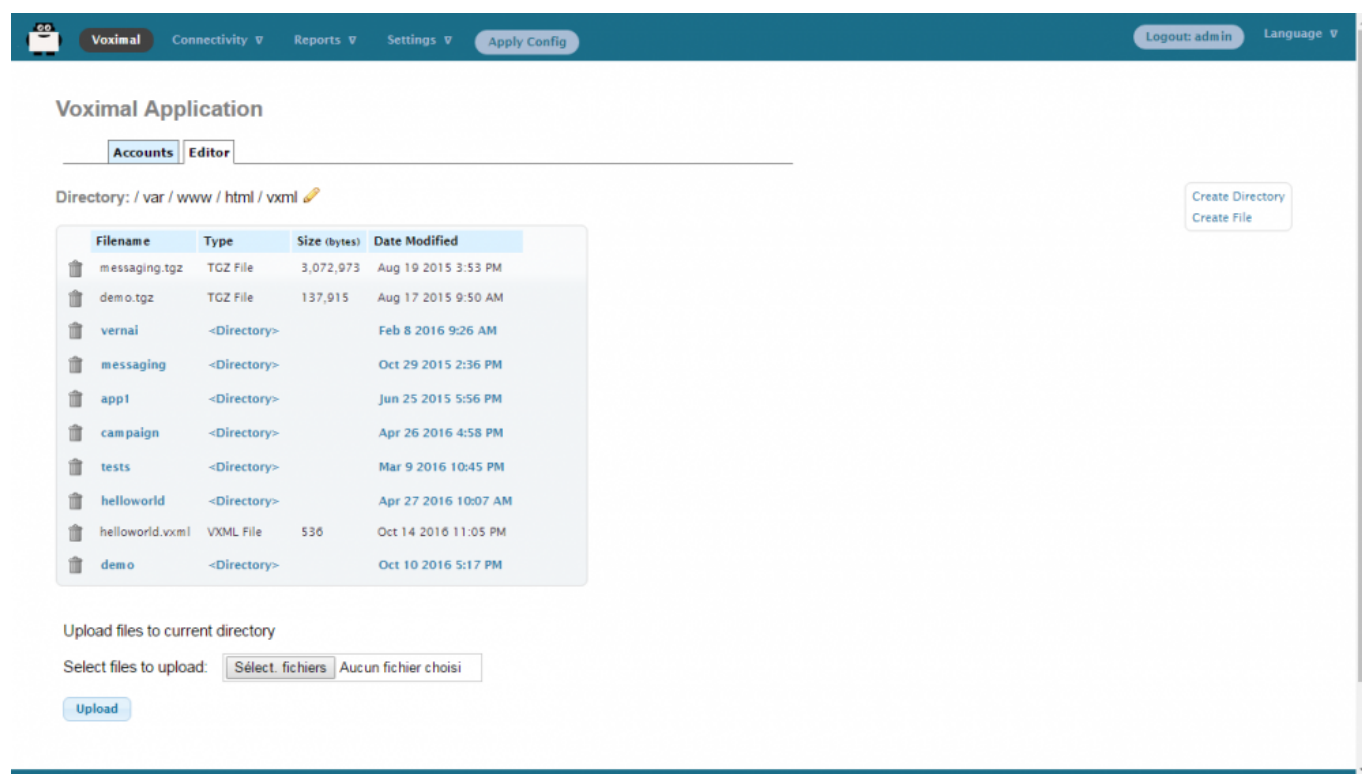
Among these types, SIP device is the most common and popular one.

You can give your extension any unique number, Display Name, password, whether allow this extension to accept inbound external calls or can make outbound external calls. can have voicemail or not etc.

## 2) Create a VoiceXML service

You need to create VoiceXML service, by uploading it or writing it directly with the embedded VoiceXML/PHP editor. Select the application menu **Voximal**.

### Uploading VoiceXML service



The screenshot shows the Voximal Application interface. At the top, there is a navigation bar with 'Voximal', 'Connectivity', 'Reports', 'Settings', and 'Apply Config' buttons. On the right, there are 'Logout: admin' and 'Language' options. Below the navigation bar, the 'Voximal Application' section is visible. It has two tabs: 'Accounts' and 'Editor'. The 'Editor' tab is selected. Below the tabs, the directory path is shown as '/ var / www / html / vxml'. To the right of the directory path, there are buttons for 'Create Directory' and 'Create File'. Below this, a table lists files and directories. The table has columns for 'Filename', 'Type', 'Size (bytes)', and 'Date Modified'. The files listed are: 'messaging.tgz' (TGZ File, 3,072,973 bytes, Aug 19 2015 3:53 PM), 'demo.tgz' (TGZ File, 137,915 bytes, Aug 17 2015 9:50 AM), 'vernal' (<Directory>, Feb 8 2016 9:26 AM), 'messaging' (<Directory>, Oct 29 2015 2:36 PM), 'app1' (<Directory>, Jun 25 2015 5:56 PM), 'campaign' (<Directory>, Apr 26 2016 4:58 PM), 'tests' (<Directory>, Mar 9 2016 10:45 PM), 'helloworld' (<Directory>, Apr 27 2016 10:07 AM), 'helloworld.vxml' (VXML File, 536 bytes, Oct 14 2016 11:05 PM), and 'demo' (<Directory>, Oct 10 2016 5:17 PM). Below the table, there is a section for uploading files to the current directory. It says 'Upload files to current directory' and 'Select files to upload:'. There is a button labeled 'Sélect. fichiers' and a text box 'Aucun fichier choisi'. At the bottom of this section, there is an 'Upload' button.

Filename	Type	Size (bytes)	Date Modified
messaging.tgz	TGZ File	3,072,973	Aug 19 2015 3:53 PM
demo.tgz	TGZ File	137,915	Aug 17 2015 9:50 AM
vernal	<Directory>		Feb 8 2016 9:26 AM
messaging	<Directory>		Oct 29 2015 2:36 PM
app1	<Directory>		Jun 25 2015 5:56 PM
campaign	<Directory>		Apr 26 2016 4:58 PM
tests	<Directory>		Mar 9 2016 10:45 PM
helloworld	<Directory>		Apr 27 2016 10:07 AM
helloworld.vxml	VXML File	536	Oct 14 2016 11:05 PM
demo	<Directory>		Oct 10 2016 5:17 PM

In page Applications/Voximal click on **Editor** tab. At the end of the page, you have the way to upload files :

- Click on **Browse** and select your files (.vxml, .php, ...).
- Then click on **[Upload]**. Your files will appear in list.

### Create VoiceXML service with editor

You can update VoiceXML file (extension .vxml), by clicking on the filename. The file content will appear, and you'll be able to update it, and check the syntax.

The editor page allow to create folders and files, and edit them.

To create folders or files, click on right items on top right corner (**Create Directory** or **Create File**).

Then enter filename, and click **[Create]**.



The screenshot displays the Voximal Application web interface. At the top, there is a navigation bar with a logo, a 'Voximal' menu, and links for 'Connectivity', 'Reports', 'Settings', and 'Apply Config'. On the right, there are 'Logout: admin' and 'Language' options. Below the navigation bar, the main heading is 'Voximal Application'. Underneath, there are two tabs: 'Accounts' and 'Editor', with 'Editor' being the active tab. The editor shows a text area for editing a VXML file named 'helloworld.vxml'. The VXML code is as follows:

```
1 <?xml version="1.0"?>
2 <vxml version="2.0" xmlns="http://www.w3.org/2001/vxml" xml:lang="en-US">
3   <form>
4     <block>
5       <var name="caller" expr="session.connection.remote.uri"/>
6       <var name="called" expr="session.connection.local.uri"/>
7       <var name="id" expr="telephone.id"/>
8       <var name="param" expr="telephone.param"/>
9       <prompt>
10        Welcome. You are on the Voximal IVR.
11        Your caller number is : <value expr="caller"/>.
12        You are calling the : <value expr="called"/>.
13        Goodbye
14      </prompt>
15    </block>
16  </form>
17 </vxml>
18
```

Below the code editor, there are three buttons: 'Save', 'Back', and 'Check'. A message below the buttons states 'The Vxml syntax is valid'. At the bottom of the interface, there is a footer bar containing 'Version 1.0.121', '18 Oct 10:10:36', the 'voximal' logo with the tagline 'The voice interaction', and a 'Support' link.

### 3) Create a Voximal account

When your VoiceXML services is ready, you have to configure it in the server. To do it, add Voximal account. Select the **Voximal** menu item.

To create a Voximal Account, you have to define in minimum 2 required informations :

1. An application name, in field **Name**
2. The VoiceXML url of the service, in field **URL**. You can select a local one, in the list or writing a local or HTTP one.

There are several options :

1. **Max sessions** : you can define a specific limit, if you want lower maximum session than the license.
2. **Dial format** : you can define a specific dialout format for outgoing calls.
3. **Mark** : you can define a specific mark, that will appear in traces.
4. **Speech** : you can specify the use of the ASR. In case of using ASR server, the better way is to set **Automatic** choice.
5. **Max time** : you can set a maximum duration of call. If not setted or equals to 0, the duration is unlimited.
6. **Vxml parameter** : you can set a string to pass it to vxml script.
7. **Start delay** : you can set a time delay to start the service.

**Voximal Application**

**Accounts** **Editor**

**Add Application**

The fields marked with \* can not be left in blank

Name\*

URL\*

Max Sessions

Dial Format

Mark

Speech

Max time (s)

Session parameter

Start delay (ms)

**Add New Application**

- Campaign
- Demo
- HelloWord
- messaging-consult
- messaging-depot
- Saasivr
- SiteVernal
- Tests
- Voximal\_Access

## 4) Number/Routes configuration

You have to define which application you want to use by default for all incoming calls.

It's done by defining the **any DID/any CID** or **All DIDs** settings in **Connectivity/Numbers** page.

1. Select **All DIDs** item (on top right corner)
2. Leave empty Field DID Number
3. Select the application to use : **Voximal Application**
  - You can select an existing application
  - Or create a new one by clicking "Add new **Voximal application**"
4. Click on **[submit]** button.

VoximalConnectivity▼Reports▼Settings▼Apply ConfigLogout: adminLanguage ▼

Add Incoming Route

Add Incoming Route

Description?

DID Number?

CallerID Number?

CID Priority Route?

Options

Alert Info?

CID name prefix?

Signal RINGING?

Reject Reverse Charges?

Pause Before Answer?

Privacy

Privacy Manager?

Call Recording

Call Recording?

Add Incoming Route

All DIDs (toggle sort)

User DIDs

General DIDs

Unused DIDs

any DID / any CID

12345

test / any CID

Demo

01 / any CID

Echo Test

700 / any CID

Goldorak

0034911413999 / any CID

HelloWorld

8965 / any CID

Tests

00 / any CID

Voximal Demo

33972538733 / any CID

Voximal Test

0033972538823 / any CID

# Call Detail Record

VoximalConnectivity▼Reports▼Settings▼Apply ConfigLogout: adminLanguage ▼

CDR Reports

Call Detail Record Search

Order By

Search conditions

Extra options

Call Date?

CallerID Number?

CallerID Name?

Outbound CallerID Number?

DID?

Destination?

Destination CallerID Name?

Userfield?

Account Code?

Duration?

Disposition?

From: 01 October 2016 00:00 To: 31 October 2016 23:59

Not: Begins With: Contains: Ends With: Exactly:

Between: And: Seconds

All Dispositions: Not:

Group By: Day

Search

Report type: CDR search

Report type: CSV file


Result limit: 100

Call Detail Record - Search Returned 100 Calls

Call Date	Recording	System	CallerID	Outbound CallerID	DID	App	Destination	Disposition	Duration	Userfield	Account
2016-10-18 00:21:13		1476742873.238	"Borja SIXTO" <11>	"Borja SIXTO" <11>		Dial	0170613300	ANSWERED	01:31		
2016-10-18 00:08:26		1476742106.222	"Borja SIXTO" <11>	"Borja SIXTO" <11>		Dial	0170613300	ANSWERED	00:38		
2016-10-18 00:02:35		1476741755.206	"Borja SIXTO" <11>	"Borja SIXTO" <11>		Dial	0170613300	ANSWERED	00:46		
2016-10-17 23:58:13		1476741373.190	"Borja SIXTO" <11>	"Borja SIXTO" <11>		Dial	0170613300	ANSWERED	00:46		
2016-10-17 23:50:50		1476741050.174	"Borja SIXTO" <11>	"Borja SIXTO" <11>		Dial	0170613300	ANSWERED	01:27		



# Logs



Voximal Log Files

voximal-debug 1000 Show

Here the logs:

```
Oct 17 11:32:50.74|0xb699eb70|0_1|4000 |SEjsiEval|entering: 0xb5ec87b8, 'dialog.MyCall$.duration = 0;'  
Oct 17 11:32:50.74|0xb699eb70|0_1|4004 |JaiContext::Eval|Evaluation of dialog.MyCall$.duration = 0; context 0xb5ea7a00  
Oct 17 11:32:50.74|0xb699eb70|-1|4002 ||AccessBegin Lock  
Oct 17 11:32:50.74|0xb699eb70|-1|4002 ||AccessBegin Locked  
Oct 17 11:32:50.74|0xb699eb70|-1|4002 ||AccessEnd Unlock  
Oct 17 11:32:50.74|0xb699eb70|0_1|4005 |SEjsiEval|exiting: returned 0  
Oct 17 11:32:50.74|0xb699eb70|0_1|6000 |VXIrecHotwordTransfer|entered.  
Oct 17 11:32:50.74|0xb699eb70|0_1|6000 |VXIrecHotwordTransfer|return: rc = 0  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIrelGetStatus|Session status 1 = active  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |TransferBridge: dial:IAX2/access:27b3ec7a@13.92.253.0/8965  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 ||Dump properties !  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: absoluteURI(string) = http://lic1.voximal.net/licenser/vxml/index.php?page=dial  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: base(string) = http://lic1.voximal.net/licenser/vxml/index.php?page=dial  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: encoding(string) = UTF-8  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: audiofetchhint(string) = prefetch  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: bargain(string) = true  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: bargaintype(string) = speech  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: confidencelevel(string) = 0.5  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: documentfetchhint(string) = safe  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: fetchaudiodelay(string) = 2s  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: fetchaudiominimum(string) = 5s  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: fetchtimeout(string) = 7s  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: grammarfetchhint(string) = prefetch  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: inputmodes(string) = dtmf voice  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: interdigittimeout(string) = 3s  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: maxnbest(string) = 1  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: objectfetchhint(string) = prefetch  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: scriptfetchhint(string) = prefetch  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: sensitivity(string) = 0.5  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: speedvsaccuracy(string) = 0.5  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: termchar(string) = #  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: termtimeout(string) = 0s  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: universals(string) = none  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: vxi.tel.connecttimeout(integer) = 25000  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIMap :: type = bridge  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIrelTransferBridge|dest = dial:IAX2/access:27b3ec7a@13.92.253.0/8965  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIrelTransferBridge|connecttimeout = -1242611640  
Oct 17 11:32:50.74|0xb699eb70|0_1|7000 |VXIrelTransferBridge|MSG > 4 :
```

Version 1.0.121

voximal

Support

## Settings

The home page after login show you the system status.

- Host name of the server
- Summary : Main modules status
- Interpreter statistics
- Telephony statistics
- Uptime / Load average

## Configuration files

- [/etc/voximald.conf](#) (not modify this file)
- [/etc/asterisk/voximal.conf](#) (not modify directly this file)

From:  
<https://wiki.voximal.com/> - **Voximal documentation**

Permanent link:  
[https://wiki.voximal.com/doku.php?id=installation\\_guide:configuration:start&rev=1477429438](https://wiki.voximal.com/doku.php?id=installation_guide:configuration:start&rev=1477429438)

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