

?????

- ?????? (?: RAM 128MB)
- OSSLab Firmware v0.3
- Asterisk 1.8.17

???????Sipp

???????sipp -sn uac -d 60000 -i 10.1.1.38 -s 7000 10.1.1.10 -l 30

???????

1. ?????? 20 ~ 38 ??????????
2. ??????????????????????
3. ??????????????? 15 ??
4. ??????? SIP Trunk ??????????????????
5. ????????????????????? IVR????????????????????IVR ??????

Web-UI > Voice Menu >

Name: sipp-test

Extension: 7000

Welcome Message: demo-instruct

t: Goto Menu > sipp-test

OpenSIPLab
http://www.opensiplab.com.tw/

Logout

System Status
Trunks
Outgoing Calling Rules
Dial Plans
Users
Ring Groups
Music On Hold
Call Queues
Voice Menu
Time Intervals
Incoming Calling Rules
Voicemail
Plaging/Intercom
Follow Me
Google Talk
Directory
Call Features
Voicemail Groups
Voice Menu Prompts
System Info
Backup
Options
Asterisk Logs
Call Detail Records
Active Channels
Bulk Add
File Editor
Asterisk CLI
AXI Settings
SIP Settings
Networking

Manage Voice Menu

Create New

Edit VoiceMenu voicemenu-custom-1

Name: sipp-test [Advanced Edit]

Extension: 7000

Allow Dialing Other Extensions

Actions

- Play demo-instrud & Listen for Key/Press events
- Wait 5 sec for the user to enter an extension

Add new Step: -- Select an Option --

- Allow Key-Press Events
- 0 --
- 1 --
- 2 --
- 3 --
- 4 --
- 5 --
- 6 --
- 7 --
- 8 --
- 9 --
- # --
- * --
- t Goto VoiceMenu -- sipp-test
- i --

Cancel Save

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GUI version: SVN branch-2.0-r5220

http://192.168.1.11/static/config/index.html

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??????? x 20

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