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- ????? firmware ?????????????? ??????????
- <http://code.google.com/p/asterisk-on-nextvod-unit/>
- Asterisk@Taiwan ??????????????????
- <https://sites.google.com/site/asterisktaiwan/>
- G+ ?? - OSSLab ??????????
- <https://plus.google.com/b/1116896284...41334496/posts>
 - ? [DIY~????????????](#)
 - ? [?? NextPBX ??????](#)
 - ? [?????? v.s ????\(????????\)](#)
 - ? [???-NextPBX ?????](#)
 - ? [????????????](#)
 - ? [????????????\(???\)](#)

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- PTT BBS ??
- <http://www.ptt.cc/bbs/NextTV/index.html>
- ?????
- <http://code.google.com/p/sh4twbox/>
- [Debian-sh4 for NextVOD](#)
 - ? <https://sites.google.com/site/debiansh4/>

NextPBX ????

IP ??????? (Voice IP PBX Features)

- ????? (IVR)
- ??????? SIP ?????????(?????)??? C2Net?IPBB?Chief070 ??????????(fail-over)???
- ????? DID ???? DID ????(?????)
- ?? SIP/IAX ???????
- ?? G711 ulaw?G711 alaw?GSM?G722?G726?iLBC ??????? (? : G722/G726 ??????)
- ????????? (?????)
- ???????
- ??????? SIP ??????ATA?VoIP Gateway? Android/iPhone ????
- ?????????????????????????
- ?????????????
- ?? [GoogleVoice](#) ??????/???
- ?? DellVoice ???????
- ?? IPBB ???
- ?? IPBB?C2Net ???? IPOX-070 ??(10?)
- ?????????????
- ??????????(Music on Hold)?????

???? (Network Features)

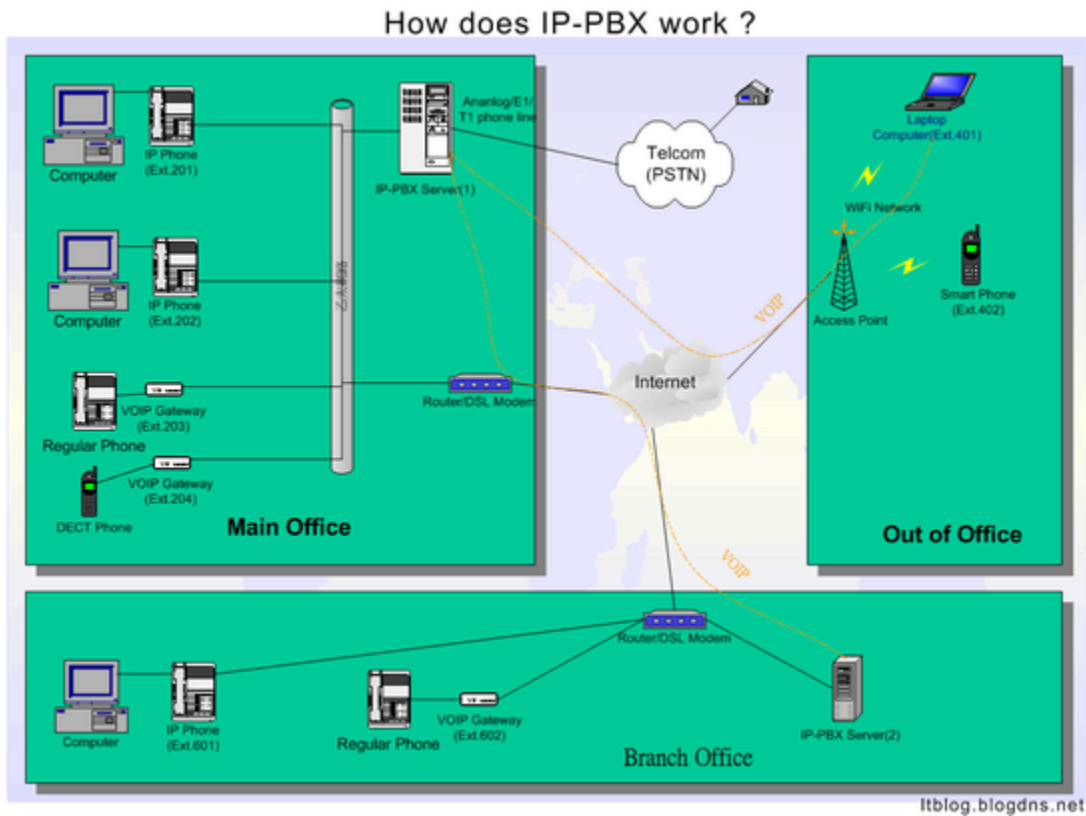
- ?? DHCP/Static/PPPoE ????

- ??????
- ????? NAT ?????

?? (Management)

- ??????? (?? Firefox/Chrome???? IE)
- ?? CDR ??????
- ?????????????

IP PBX ????????



?? IP PBX ???

- <http://www.ipbbnet.com/voip/td04.asp>
- http://www.ggutc.com.tw/NEWS/news_ippbx_upg.htm
- <http://www.tonesky.com/services-ip-pbx.php>
- <http://www.ithome.com.tw/itadm/article.php?c=57444>
- <http://www.netcom.com.tw/Mosa/index.htm>

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- STLinux 2.3
- Asterisk 1.8.17
- Asterisk GUI 2.0 revision 5520

- iptables

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- CPU: STMicroelectronics [ST-9150BUC](#) (Renesas SuperH Core)
- Ethernet: VIA VT6113 (Ethernet)
- RAM: NANYA NT5TU64M16GG-AC (DRAM DDR2 64Mx16 PC800 BGA) * 2= 256MBytes (???????????)
~~128MB~~)
- Storage: Samsung NCDOM08GMS-MLC (USB DOM 8GB)
- Winbond 25Q80 1MB Spi Flash (Onboard Flash)
- ??????????????????
? [?????????????](#)
? [T??](#)

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- ??????pdk7105
- ?????????????????

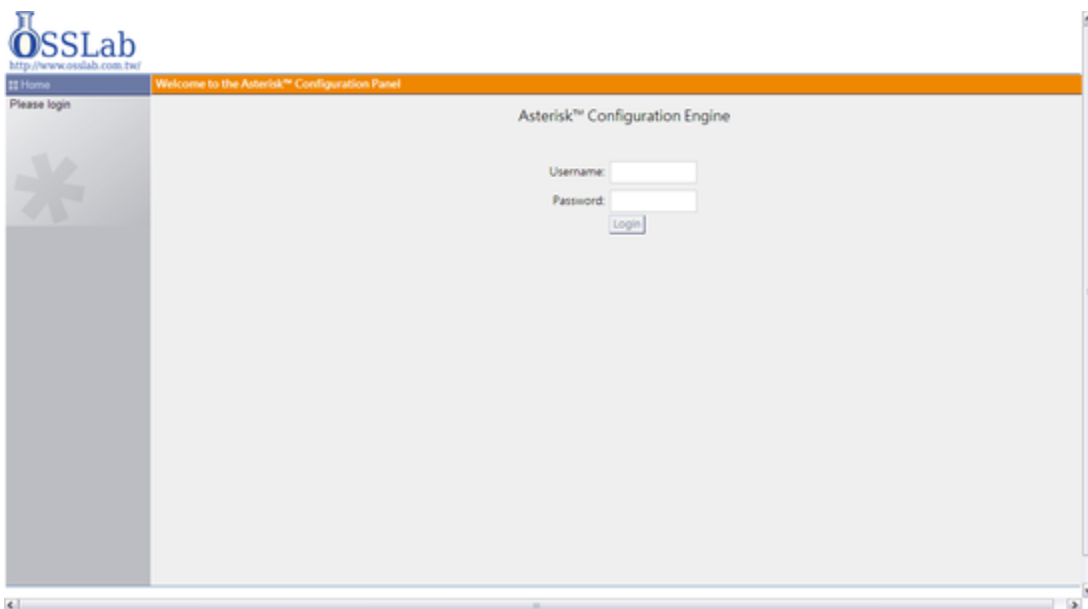
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- ToDoList: <http://code.google.com/p/asterisk-on.../wiki/ToDoList>
- ChangeLogs: <http://code.google.com/p/asterisk-on...iki/ChangeLogs>

NextPBX ???????

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System Status

Please click on a panel to manage related features

Trunks

Status	Trunk	Type	Username	Port/Hostname/IP
Registered	ipbb	sip	0702180150	msw1.mkyc.com

Extensions

Legend: Free (Green), Ringing (Yellow), Busy (Red), UnAvailable (Grey)

Extension	Name/Label	Status	Type
6000	alang	Messages: 1/1	SIP User
-- *No Extension assigned			
Check Voicemails			
VoiceMailMain			
-- *No Extension assigned			
Dial by Names			
Directory			

Queues

Parking Lot

Caller ID	Channel	Extension	Timeout
No Parked Calls			

System Info

General Network Memory Disk Hostname

localhost

OS Version:
Linux localhost 2.6.23.17_#1 SMP Fri Aug 10 22:03:09 EDT 2007 i686 GNU/Linux

NextPBX Version:
NextPBX 0.6.3 created by A.Lang@OSSLab

Asterisk Build:
Asterisk/1.8.17.0
Asterisk GUI-version : SVN-branch-2.0-r2520

Server Date & Timezone:
Sun Dec 23 09:06:43 CST 2012

Uptime:
09:06:42 up 18:25,
Load Average: 0.93, 0.28, 0.10

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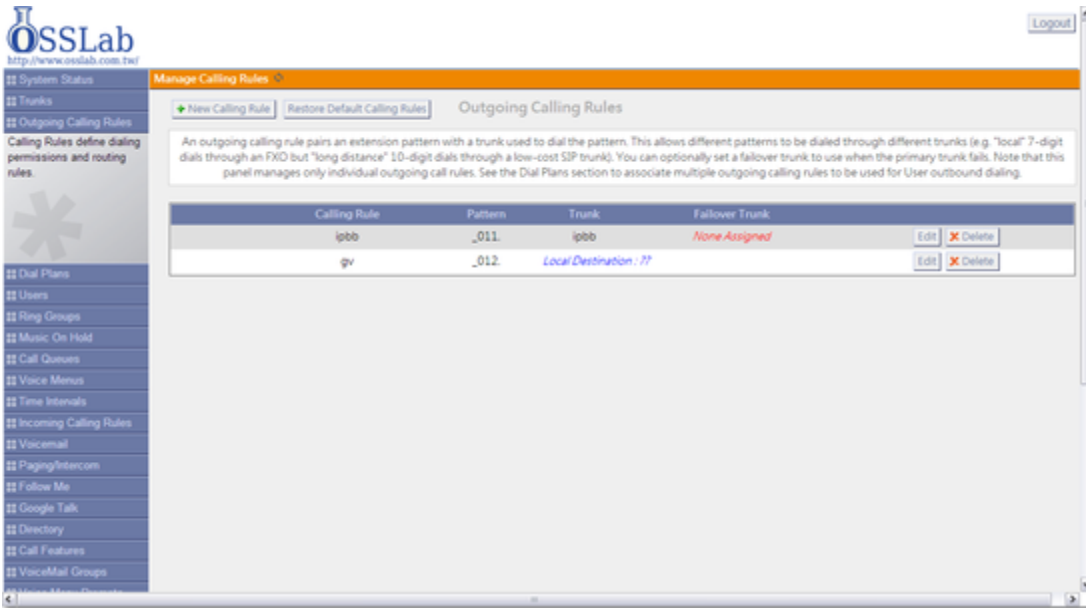
Manage SIP & IAX trunks

Analog Trunks **VOIP Trunks** T1/E1/ISDN Trunks

+ New SIP/IAX Trunk

Provider Name	Type	Hostname/IP	Username
ipbb	SIP	msw1.mkyc.com	0702180150

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OSSLab
http://www.osslab.com.tw

System Status | **Manage Calling Rules** | Logout

Trunks | Outgoing Calling Rules | Dial Plans | Users | Ring Groups | Music On Hold | Call Queues | Voice Menus | Time Intervals | Incoming Calling Rules | Voicemail | Paging/Intercom | Follow Me | Google Talk | Directory | Call Features | VoiceMail Groups

Calling Rules define dialing permissions and routing rules.

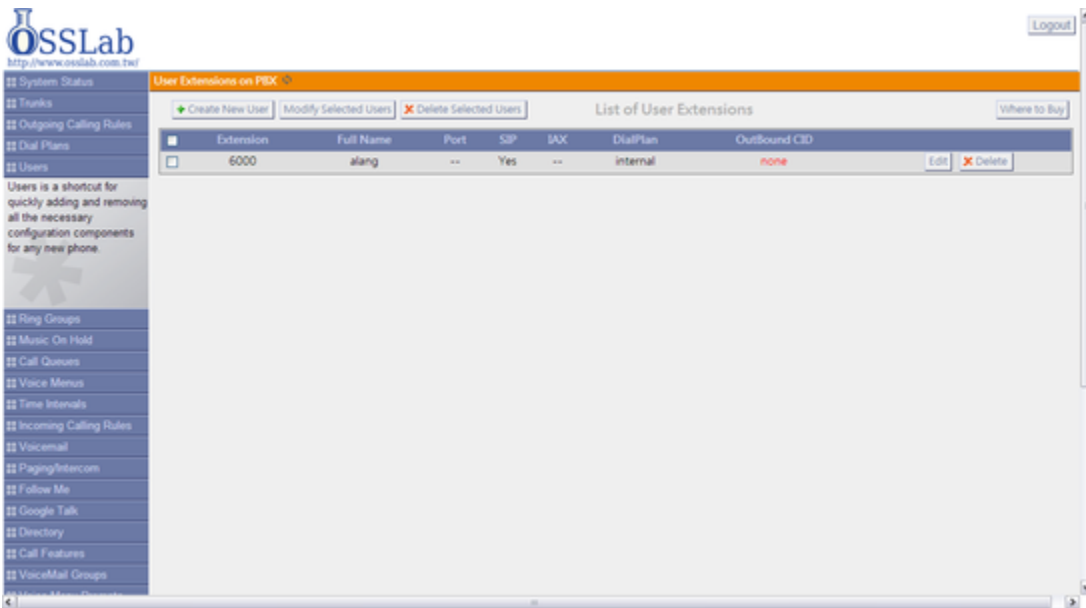
Outgoing Calling Rules

+ New Calling Rule | Restore Default Calling Rules

An outgoing calling rule pairs an extension pattern with a trunk used to dial the pattern. This allows different patterns to be dialed through different trunks (e.g. "local" 7-digit dials through an FXO but "long distance" 10-digit dials through a low-cost SIP trunk). You can optionally set a failover trunk to use when the primary trunk fails. Note that this panel manages only individual outgoing call rules. See the Dial Plans section to associate multiple outgoing calling rules to be used for User outbound dialing.

Calling Rule	Pattern	Trunk	Fallover Trunk	
ipbb	_011	ipbb	None Assigned	Edit Delete
gv	_012	Local Destination : ??		Edit Delete

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OSSLab
http://www.osslab.com.tw

System Status | **User Extensions on PBX** | Logout

Trunks | Outgoing Calling Rules | Dial Plans | **Users** | Ring Groups | Music On Hold | Call Queues | Voice Menus | Time Intervals | Incoming Calling Rules | Voicemail | Paging/Intercom | Follow Me | Google Talk | Directory | Call Features | VoiceMail Groups

Users is a shortcut for quickly adding and removing all the necessary configuration components for any new phone.

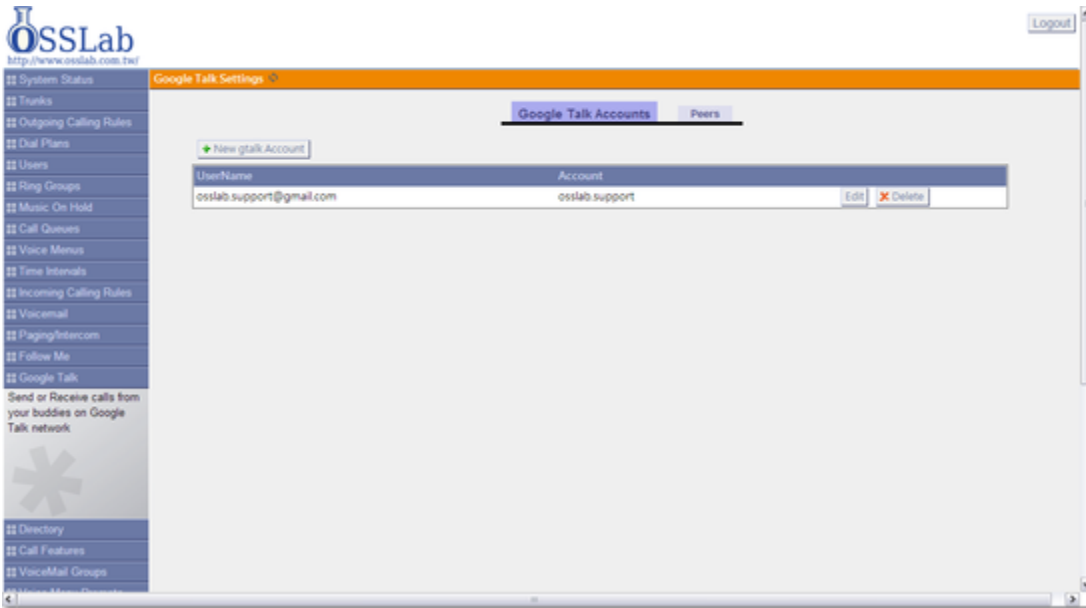
User Extensions on PBX

+ Create New User | Modify Selected Users | Delete Selected Users | Where to Buy

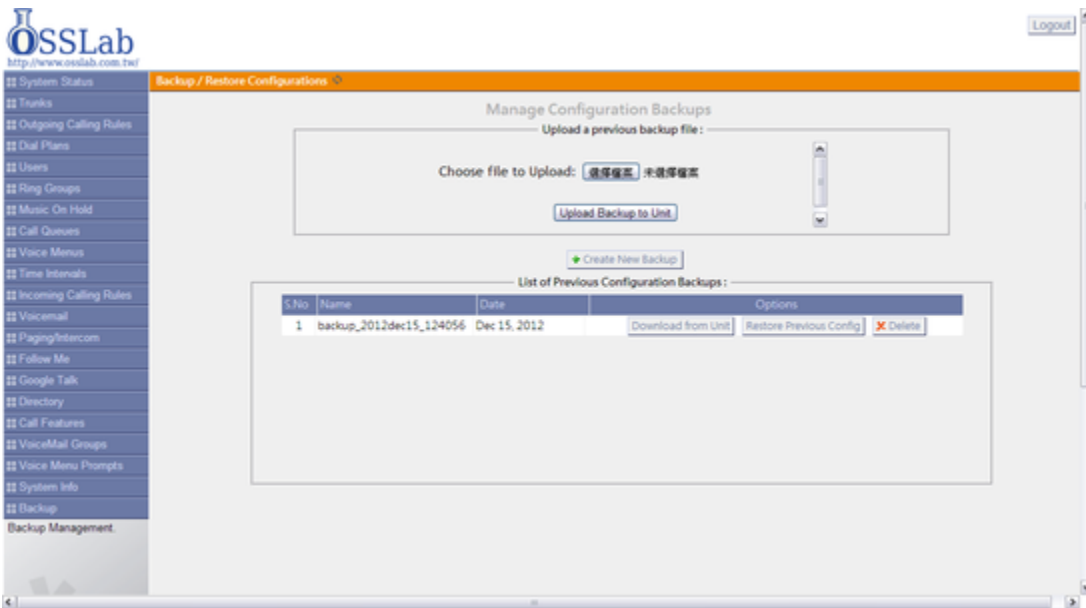
List of User Extensions

Extension	Full Name	Port	SIP	SAX	DialPlan	Outbound CID	
<input type="checkbox"/> 6000	alang	--	Yes	--	internal	none	Edit Delete

Google Voice



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System Status

Trunks

Outgoing Calling Rules

Dial Plans

Users

Ring Groups

Music On Hold

Call Queues

Voice Menus

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 **beta**

System Status **beta**

Networking setting

Network Interface

DHCP: yes

Hostname:

Domain:

IP address:

Subnet mask:

Gateway:

DNS:

NTP:

System TimeZone

TimeZone: