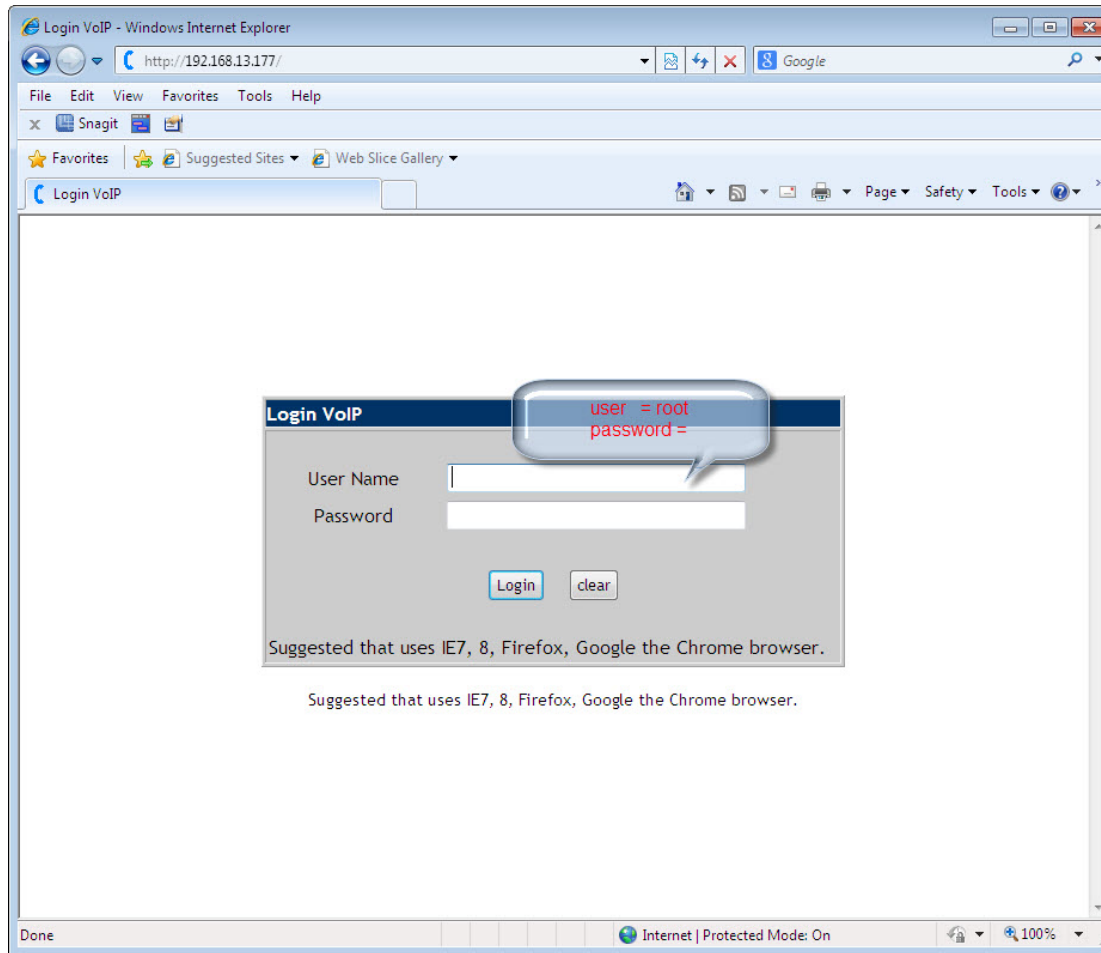


VIP-157S




Login

(default Lan fix ip = 192.168.0.1 , PC = bridge)



General setting

 **PLANET**
Networking & Communication

**VoIP Phone Adapter
Configuration Menu**

- Status
- Wizard
- Phone Book
- Phone Settings**
- Call Service
- SNTP Settings
- Volume Settings
- Dial Plan - Basic
- Dial Plan - Advanced
- General**
- Network
- NAT Trans
- SIP Settings
- Advanced Settings
- System Auth
- System Settings
- Save and Reboot
- Logout

General Setting

You could set the general options of your phone in this page.

Call Waiting:	Disable ▾
Ring Timeout:	60 ▾ (seconds)
Caller ID Scheme:	FSK (Bellcore) ▾
CID Type II:	Disable ▾
T.38 (FAX):	Disable ▾
T.38 Pass-Through Codec:	aLaw ▾

Network setting



VoIP Phone Adapter Configuration Menu

Status

Wizard

Phone Book

Phone Settings

Network

Network Settings

DDNS Settings

VLAN

VPN (PPTP & L2TP)

IPv6

NAT Trans

SIP Settings

Advanced Settings

System Auth

System Settings

Save and Reboot

Logout

WAN Setting

You could configure the Network settings in this page.

WAN Active:	<input type="text" value="DHCP"/>
IP Address:	<input type="text" value="192.168.13.177"/>
Subnet Mask:	<input type="text" value="255.255.255.0"/>
Default Gateway:	<input type="text" value="192.168.13.1"/>
DNS Active:	<input type="text" value="Static"/>
Primary DNS:	<input type="text" value="168.95.192.1"/>
Second DNS:	<input type="text" value="168.95.1.1"/>
MAC Address:	<input type="text" value="00:30:4f:9f:79:2b"/>
System Name:	<input type="text" value="VOIP_TA25"/>

Sip Account



VoIP Phone Adapter Configuration Menu

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SIP Settings

Service Domain

Codecs Settings

Advanced

Advanced Settings

System Auth

System Settings

Save and Reboot

Logout

Service Domain Setting

You could set the configuration of Service Domain in this page.

Phone: 1

Realm: 1

Realm Active:	Enable
Display Name:	021053008
Phone Number:	+6621053008
Authentication ID:	6621053008@catnextgen.com
Authentication Password:
Domain Server:	catnextgen.com
Proxy Server:	202.129.61.102
Outbound Proxy:	202.129.61.102
Subscribe for MWI :	Disable

Submit

Reset

Set Codecs



VoIP Phone Adapter Configuration Menu

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Service Domain

Codecs Settings

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Advanced Settings

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System Settings

Save and Reboot

Logout

Codecs Setting

You could set the configuration of Codec in this page.

Disable Codecs		Enable Codecs
G.711 u-law G.726 - 16 G.726 - 24 G.726 - 32 G.726 - 40 G.723	>> <<	G.729 G.711 a-law
	Move	MoveUp MoveDown

G.711 and G.729:	20	ms
G.723:	30	ms
G.723 5.3K:	Disable	
Silence Suppression (VAD):	Disable	
Echo Canceller :	Enable	

Codec Type		ID Value
G726-16:	Default	23 (95-127)
G726-24:	Default	22 (95-127)
G726-32:	Default	2 (95-127)
G726-40:	Default	21 (95-127)
RFC 2833:	Custom	101 (95-127)

Set DTMF

The screenshot displays a web browser window with the URL <http://192.168.13.177/>. The page title is "SIP - Advanced Setting" and it is part of the "PLANET Networking & Communication" interface. A sidebar on the left contains a navigation menu with items like "Status", "Wizard", "Phone Book", "Phone Settings", "Network", "NAT Trans", "SIP Settings", "Service Domain", "Codecs Settings", "Advanced", "Advanced Settings", "System Auth", "System Settings", "Save and Reboot", and "Logout".

The main content area shows a configuration table for SIP settings. The "DTMF Type" dropdown menu is open, showing the following options: RFC 2833, Inband, SIP Info, and RFC 2833 + Inband (which is highlighted with a red box). The "SIP QoS (Diff-Serv)" dropdown menu is also open, showing the following options: RFC 2833 + Inband (highlighted with a red box), SIP Info + Inband, and (0-255).

SIP Expire Time:	60	(60-86400 Seconds, 0=define by Server)
SIP Expire Time Type:	General	((General: Expire Time - [Expire Time]/6))
SIP Registration Retry Timer:	64	(5-250 Seconds)
SIP Session Timer T1:	500	(ms)
SIP Session Timer T2:	4000	(ms)
SIP Session Timer B, F, H:	32000	(ms)
SIP INVITE Timeout:	32000	(ms)
Local SIP Port of Phone 1:	5060	~ 5060 (1024-40000, Start - End)
Local RTP Port of Phone 1:	20000	~ 21999 (1024-40000, Start - End)
Local SIP Port of Phone 2:	5062	~ 5062 (1024-40000, Start - End)
Local RTP Port of Phone 2:	22000	~ 23999 (1024-40000, Start - End)
Hold Type:	RFC 2543 (0.0.0.0)	
DTMF Type:	RFC 2833	
RPort:	RFC 2833	
Voice QoS (Diff-Serv):	Inband	
SIP QoS (Diff-Serv):	RFC 2833 + Inband	
RTP Traffic Class (IPv6):	46	(0-255)
SIP Traffic Class (IPv6):	40	(0-255)
Use DNS SRV:	Disable	
Keep-alive Message:	Disable	
Keep-alive Interval:	60	(15-250 Seconds)
Jitter Buffer:	1	~ 32 (1-32 Packet)

Save Config



VoIP Phone Adapter Configuration Menu

Status

Wizard

Phone Book

Phone Settings

Network

NAT Trans

SIP Settings

Advanced Settings

System Auth

System Settings

Firmware Update

Auto Update

Reset to Default

Save and Reboot

Logout

Save and Reboot System

You have to save changes to effect them.

Save Change:

Save

Up firmware (กรณีใช้ใน NGN CAT)

VoIP Web Management - Windows Internet Explorer

http://192.168.1.12/update.htm

File Edit View Favorites Tools Help

VoIP Web Management

PLANET
Networking & Communication

VoIP Phone Adapter Configuration Menu

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System Settings
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Logout

Update System

Update Active: ALL ROM xxxx.rom

Load From File:

<http://x.x.x.x/update.htm>
ต้องใช้ url ถึงจะสามารถ upfirmwareได้

Internet | Protected Mode: On 100%

11:47 29/11/2556