



- Account 1 ▶
- Account 2 ▶

- Accounts
- Account 1 -
- General Settings
- Network Settings
- SIP Settings +
- Audio Settings
- Call Settings
- Account 2 +

General Settings

Account Active	<input type="radio"/> No <input checked="" type="radio"/> Yes
Account Name	<input type="text" value="#####"/>
SIP Server	<input type="text" value="sp.voipus.net"/>
Secondary SIP Server	<input type="text"/>
Outbound Proxy	<input type="text"/>
Backup Outbound Proxy	<input type="text"/>
SIP User ID	<input type="text" value="#####"/>
Authenticate ID	<input type="text" value="#####"/>
Authenticate Password	<input type="text"/>
Name	<input type="text" value="#####"/>
Voice Mail UserID	<input type="text" value="#####"/>



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Network Settings

DNS Mode

A Record

Primary IP

Backup IP 1

Backup IP 2

NAT Traversal

Keep-Alive

Proxy-Require

Save

Save and Apply

Reset



Accounts

Account 1



General Settings

Network Settings

SIP Settings



Audio Settings

Call Settings

Account 2



Audio Settings

Send DTMF

 in-audio via RTP (RFC2833) via SIP INFO

DTMF Payload Type

101

Preferred Vocoder - choice 1

iLBC

Preferred Vocoder - choice 2

PCMU

Preferred Vocoder - choice 3

PCMA

Preferred Vocoder - choice 4

G.729A/B

Preferred Vocoder - choice 5

G.723.1

Preferred Vocoder - choice 6

G.726-32

Preferred Vocoder - choice 7

G.722(wide band)

Use First Matching Vocoder in 2000K SDP

 No Yes

SRTP Mode

No

Symmetric RTP

 No Yes

Silence Suppression

 No Yes

Voice Frames per TX

2

G723 Rate

 6.3kbps encoding rate 5.3kbps encoding rate

G.726-32 Packing Mode

 ITU IETF

iLBC Frame Size 20ms 30ms

iLBC Payload Type

Jitter Buffer Type

Jitter Buffer Length

Save

Save and Apply

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Accounts

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Call Settings

Early Dial No Yes

Dial Plan Prefix

Dial Plan { x+ | *x+ | *xx*x+ | xx

Delayed Call Forward Wait Time

Enable Call Features No Yes

Dial Plan設定変更: { x+ | *x+ | *xx*x+ } ---> { x+ | *x+ | *xx*x+ | xxx*x | xxxxxx*x | x** }

ntp-a3.nict.go.jp



Settings

General Settings

Call Features

Multicast Paging

Ring Tone

Audio Control

LCD Display

Date and Time

Web Service

XML Applications

Programmable Keys

Date and Time

NTP Server

Allow DHCP Option 42 to override
NTP server No Yes

Time Zone

 Allow DHCP Option 2 to Override Time Zone Sett

Self-Defined Time Zone

Date Display Format

Time Display Format

 12 Hour 24 Hour

Save

Save and Apply

Reset

NTP Server

Defines the URL or IP address of the NTP server. The phone may obtain the date and time from the server.

[Reset to Default](#)[Undo](#)



Settings

General Settings

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Programmable Keys

Local RTP Port

Use Random Port No Yes

Keep-Alive Interval

Use NAT IP

STUN server

Public Mode No Yes

Public Mode Login Expiration

Save Save and Apply Reset

Auto firmware.grandstream.com

Farmware Download: wats.gr.jp/downloads/GrandStream/Release_GXP1450_1.0.8.6/