

Telefon se konfiguruje přes www rozhraní. Po zapojení podle ukázkového zapojení zvedněte sluchátko a stiskněte tlačítko menu na telefonu. Na displeji se zobrazí IP adresa telefonu. Tuto adresu zadejte do www prohlížeče na počítači. (co tři číslice zadejte "." např.:192.168.1.123). Původní heslo je "admin".

Sip Server a Outbound Proxy: zjistíte na <http://www.802.cz/zjistení-sip-serveru/>

Grandstream Device Configuration		
STATUS	BASIC SETTINGS	ADVANCED SETTINGS
MAC Address:	00.0B.82.02.C0.96	
WAN IP Address:		
Product Model:	BT100	
Software Version:	Program-- 1.0.6.7 Bootloader-- 1.0.1.0 HTML-- 1.0.0.49 VOC-- 1.0.1.0	
System Up Time:	0 day(s) 0 hour(s) 9 minute(s)	
Registered:	No	
PPPoE Link Up:	disabled	
NAT:	detected NAT type is open Internet	
NAT Mapped IP:	0.0.0.0	
NAT Mapped Port:	0	
Total Inbound Calls:	0	
Total Outbound Calls:	0	
Total Missed Calls:	0	
Total Call Time (in minutes):	0	
Total SIP Message Sent:	0	
Total SIP Message Received:	0	
Total RTP Packet Sent:	0	
Total RTP Packet Received:	0	
Total RTP Packet Loss:	0	

Grandstream Device Configuration

STATUS

BASIC SETTINGS

ADVANCED SETTINGS

End User Password: (purposely not displayed for security protection)

IP Address: dynamically assigned via DHCP (default) or PPPoE
(will attempt PPPoE if DHCP fails and following is non-blank)

PPPoE account ID:

PPPoE password:

Preferred DNS server: . . .

statically configured as:

IP Address: . . .

Subnet Mask: . . .

Default Router: . . .

DNS Server 1: . . .

DNS Server 2: . . .

Time Zone: ▼

Daylight Savings Time: No Yes (if set to Yes, display time will be 1 hour ahead of normal time)

Year-Month-Day

Date Display Format: Month-Day-Year

Day-Month-Year

Update

Cancel

Reboot

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Dial Plan Prefix: (this prefix string is added to each dialed number)

No Key Entry Timeout: (in seconds, default is 4 seconds)

Use # as Dial Key: No Yes (if set to Yes, "#" will function as the "(Re-)Dial" key)

local SIP port: (default 5060)

local RTP port: (1024-65535, default 5004)

Use random port: No Yes

NAT Traversal: No
 Yes, STUN server is: (URI or IP:port)

keep-alive interval: (in seconds, default 20 seconds)

Use NAT IP: (if specified, this IP address is used in SIP/SDP message)

Proxy-Require: (if specified, the content will appear in Proxy-Require header)

Firmware Upgrade: Via TFTP Server . . .
 Via HTTP Server
Automatic HTTP Upgrade:
 No Yes, check for upgrade every days (default 7 days)

Voice Mail UserID: (User ID/extension for 3rd party voice mail system)

SUBSCRIBE for MWI: No, do not send SUBSCRIBE for Message Waiting Indication
 Yes, send periodical SUBSCRIBE for Message Waiting Indication

Auto Answer: No Yes

Offhook Auto-Dial: (User ID/extension to dial automatically when offhook)

Enable Call Features: No Yes (if Yes, Call Forwarding & Call-Waiting-Disable are supported locally)

Disable Call-Waiting: No Yes

Send DTMF: in-audio via RTP (RFC2833) via SIP INFO

DTMF Payload Type:

Send Flash Event: No Yes (Flash will be sent as a DTMF event if set to Yes)

NTP Server: (URI or IP address)

Default Ring Tone: system ring tone
 custom ring tone 1, used if incoming caller ID is
 custom ring tone 2, used if incoming caller ID is
 custom ring tone 3, used if incoming caller ID is

Send Anonymous: No Yes (caller ID will be blocked if set to Yes)

Lock keypad update: No Yes (configuration update via keypad is disabled if set to Yes)

Syslog Server:

Syslog Level: