

Brána se konfiguruje přes www rozhraní a pomocí klávesnice připojeného telefonu. Původní heslo je "admin".

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

Grandstream Device Configuration		
STATUS	BASIC SETTINGS	ADVANCED SETTINGS
MAC Address:	00.0B.82.04.2B.2B	
WAN IP Address:		
Product Model:	HT487	
Software Version:	Program-- 1.0.6.7 Bootloader-- 1.0.1.0 HTML-- 1.0.0.49 VOC-- 1.0.0.10	
System Up Time:	1 day(s) 3 hour(s) 1 minute(s)	
Registered:	Yes	
PPPoE Link Up:	disabled	
NAT:		
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:	29	
Total SIP Message Received:	59	
Total RTP Packet Sent:	0	
Total RTP Packet Received:	0	
Total RTP Packet Loss:	0	

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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: GMT+1:00 (Paris, Amsterdam, Berlin, Rome, Vienna, Madrid, Warsaw, Brussels)

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

NAT/Bridge/DHCP Server Information & Configuration:

Device Mode: NAT Router Bridge

Cloned WAN MAC Addr: (in hex format)

LAN Subnet Mask: (default is 255.255.255.0)

LAN DHCP Base IP: (base IP for the LAN port, default is 192.168.2.1)

DHCP IP Lease Time: (in units of hours, default is 120 hours or 5 days)

DMZ IP:

Port Forwarding:	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>
	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>
	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>
	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>
	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>
	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>
	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>
	WAN port	<input type="text"/>	LAN IP	<input type="text"/>	LAN port	<input type="text"/>	Protocol	"UDP Only" <input type="text"/>

Update

Cancel

Reboot

Grandstream Device Configuration

STATUS
BASIC SETTINGS
ADVANCED SETTINGS

Admin Password: (purposely not displayed for security protection)
SIP Server: (e.g., sip.mycompany.com, or IP address)
Outbound Proxy: (e.g., proxy.myprovider.com, or IP address, if any)
SIP User ID: (the user part of an SIP address)
Authenticate ID: (can be identical to or different from **SIP User ID**)
Authenticate Password: (purposely not displayed for security protection)
Name: (optional, e.g., John Doe)

Advanced Options:

Preferred Vocoder: choice 1:
 (in listed order) choice 2:
 choice 3:
 choice 4:
 choice 5:
 choice 6:
 choice 7:

G723 rate: 6.3kbps encoding rate 5.3kbps encoding rate

iLBC frame size: 20ms 30ms

iLBC payload type: (between 96 and 127, default is 97)

Silence Suppression: No Yes

Voice Frames per TX: (up to 10/20/32/64 for G711/G726/G723/other codecs respectively)

Fax Mode: T.38 (Auto Detect) Pass-Through

Layer 3 QoS: (Diff-Serv or Precedence value)

Layer 2 QoS (VoIP): 802.1Q/VLAN Tag 802.1p priority value (0-7)

Layer 2 QoS (PC): 802.1Q/VLAN Tag 802.1p priority value (0-7)

Use DNS SRV: No Yes

User ID is phone number: No Yes

SIP Registration: Yes No

Unregister On Reboot: Yes No

Register Expiration: (in minutes. default 1 hour, max 45 days)

Early Dial: No Yes (use "Yes" only if proxy supports 484 response)

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)

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

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)

FXS Impedance:

Caller ID Scheme:

Onhook Voltage:

Polarity Reversal: No Yes (reverse polarity upon call establishment and termination)

NTP Server: (URI or IP address)

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)

Special Feature:

WAN Side HTTP Access: No Yes (WAN side access to http server will be rejected if set to No)

PSTN Access Code: (key pattern to use the PSTN line, default is "*00")

Syslog Server:

Syslog Level: