

Grandstream UCM6308 IP PBX

Product Name: Grandstream UCM6308 IP PBX

Manufacturer: Grandstream

Model Number: UCM6308

Grandstream UCM6308 IP PBX

The Grandstream UCM6308 allows businesses to build powerful and scalable unified communication and collaboration solutions. This series of IP PBXs provide a platform that unifies all business communication on one centralised network, including voice, video calling, video conferencing, video surveillance, web meetings, data, analytics, mobility, facility access, intercoms and more. The UCM6300 series supports up to 3000 users and includes a built-in web meetings and video conferencing solution that allows employees to connect from the desktop, mobile, GVC series devices and IP phones. It can be paired with the UCM6300 ecosystem to offer a hybrid platform that combines the control of an on-premise IP PBX with the remote access of a cloud solution.

Grandstream UCM6308 Key Features

- Supports up to 3000 users and up to 450 concurrent calls
- Zero configuration provisioning of Grandstream SIP endpoints
- Built-in conferencing & meetings platform; supports desktop, Wave app, and SIP endpoints
- Wave for Android, iOS, Chrome and Firefox browsers allows communication with all UCM6300 users & solutions
- API available for third-party integrations, including CRM and PMS platforms
- Advanced security protection with secure boot, unique certificate and random default password to protect calls and accounts
- Three Gigabit auto-sensing RJ45 network ports with integrated PoE+ and support NAT router
- Automated NAT firewall traversal service facilitates secure remote connections
- Supports Full-Band Opus voice codec and H.264/H.263/ H.263+/H.265/VP8 video codec, jitter resilience up to 50% packet loss
- Compatible with GDMS for cloud setup, management and monitoring
- Based on Asterisk* version 16 open source telephony operating system

The UCM6300 ecosystem consists of the Wave app for web and mobile, which provides a hub for collaborating remotely, and UCM RemoteConnect, a cloud NAT traversal service for ensuring secure remote connections. The UCM6300 series also offers cloud setup and management through GDMS and an API for integration with third-party platforms. By offering a high-end unified communications and collaboration solution packed with a suite of mobility, security, meeting and collaboration tools, the UCM6300 series provides a powerful platform for any organisation.

Grandstream UCM6308 - Technical Specifications

Analog Telephone FXS Ports

• 8 RJ11 Port

• All ports have lifeline capability in case of power outage

PSTN Line FXO Ports

• 8 RJ11 Port

• All ports have lifeline capability in case of power outage

Network Interfaces

• Three self-adaptive Gigabit ports (switched, routed or dual mode) with PoE+

NAT Router

Grandstream UCM6308 IP PBX

• Yes (supports router mode and switch mode)

Peripheral Ports

• 2*USB 3.0

• 1*SD card interface

LED Indicators

• Power 1/2

• FXS

• FXO

• LAN

• WAN

• Heartbeat

LCD Display

• 128x32 dot matrix graphic LCD with DOWN and OK buttons

Reset Switch

• Yes, long press for factory reset and short press for reboot

Voice-over-Packet Capabilities

• LEC with NLP Packetized Voice Protocol Unit

• 128ms-tail-length carrier grade Line Echo Cancellation

• Dynamic Jitter Buffer

• Modem detection & auto-switch to G.711

• NetEQ

• FEC 2.0

• Jitter resilience up to 50% audio packet loss

Voice and Fax Codecs

• Opus, G.711 A-law/U-law, G.722, G.722.1 G.722.1C, G.723.1 5.3K/6.3K, G.726-32, G.729A/B, iLBC, GSM; T.38

Video Codecs

• H.264, H.263, H.263+, H.265, VP8

QoS

• Layer 2 QoS (802.1Q, 802.1p) and Layer 3 (ToS, DiffServ, MPLS) QoS

API

• Full API available for third-party platform and application integration

Telephony Operating System

• Based on Asterisk version 16

Grandstream UCM6308 IP PBX

DTMF Methods

• In-band audio, RFC2833, and SIP INFO

Provisioning Protocol & Plug-and-Play

• Mass provisioning using AES encrypted XML configuration file, auto-discovery & auto-provisioning of Grandstream IP endpoints via ZeroConfig (DHCP Option 66 multicast SIP SUBSCRIBE mDNS), eventlist between local and remote trunk

Network Protocols

• TCP/UDP/IP, RTP/RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP, SSH, HTTP/HTTPS, PPPoE, STUN, SRTP, TLS, LDAP, HDLC, HDLC-ETH, PPP, Frame Relay (pending), IPv6, OpenVPN®

Disconnect Methods

• Busy/Congestion/Howl Tone, Polarity Reversal, Hook Flash Timing, Loop Current Disconnect

Media Encryption

• SRTP, TLS, HTTPS, SSH, 802.1X

Universal Power Supply

• 2x DC 12V Power Jack
• Input: 100~240VAC, 50/60Hz
• Output: DC 12V, 2A

Dimensions

• 485mm(L) x 187.2mm(W) x 46.2mm(H)

Weight

• Unit Weight: 2550g
• Package Weight: 3320g

Temperature & Humidity

• Operating: 32 - 113°F / 0 ~ 45°C, Humidity 10 - 90% (non-condensing)
• Storage: 14 - 140°F / -10 ~ 60°C, Humidity 10 - 90% (non-condensing)

Mounting

• Rack mount & Desktop

Multi-Language Support

• Web UI: English, Simplified Chinese, Traditional Chinese, Spanish, French, Portuguese, German, Russian, Italian, Polish, Czech, Turkish
• Customisable IVR/voice prompts: English, Chinese, British English, German, Spanish, Greek, French, Italian, Dutch, Polish, Portuguese, Russian, Swedish, Turkish, Hebrew, Arabic,

Grandstream UCM6308 IP PBX

Nederlands

ï½ Customisable language pack to support any other languages

Caller ID

ï½ Bellcore/Telcordia, ETSI-FSK, ETSI-DTMF, SIN 227 ‐ BT, NTT

Polarity Reversal/Wink

ï½ Yes, with enable/disable option upon call establishment and termination

Call Center

ï½ Multiple configurable call queues, automatic call distribution (ACD) based on agent skills/availability/ work-load, in-queue announcement

Customisable Auto Attendant

ï½ Up to 5 layers of IVR (Interactive Voice Response) in multiple languages

Maximum Call Capacity

ï½ Users: 3000

ï½ Concurrent calls (G.711): 450

ï½ Max concurrent SRTP calls (G.711): 300

Maximum Attendees of Conference Bridges

ï½ 8 Video Conference rooms and up to 60 parties with 1080p, assuming 4 video feeds + 1 screen sharing (H.264 & G.711)

ï½ Voice Conference: Up to 300 parties (G.711)

Wave Mobile App

ï½ Allows Android & iOS users to join UCM-hosted meetings & communicate with other users/solutions registered to the UCM6300

Call Features

ï½ Call park, call forward, call transfer, call waiting, caller ID, call record, call history, ringtone, IVR, music on hold, call routes, DID, DOD, DND, DISA, ring group, ring simultaneously, time schedule, PIN groups, call queue, pickup group, paging/intercom, voicemail, call wakeup, SCA, BLF, voicemail to email, fax to email, speed dial, call back, dial by name, emergency call, call follow me, blacklist/whitelist, voice conference, video conference, eventlist, feature codes, busy camp-on/ call completion, voice control

Firmware Upgrade

ï½ Supported by Grandstream Device Management System (GDMS), a zero-touch cloud provisioning and management system, It provides a centralized interface to provision, manage, monitor and troubleshoot Grandstream products

Compliance

ï½ FCC: Part 15 (CFR 47) Class B, Part 68

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CE: EN 55032, EN 55035, EN 61000-3-2, EN 61000-3-3, EN 62368-1, ETSI ES 203 021, ITU-T K.21

IC: ICES-003, CS-03 Part I Issue 9

RCM: AS/NZS CISPR 32, AS/NZS 62368.1, AS/CA S002, AS/CA S003.1/.2

Power adapter: UL 60950-1 or UL 62368-1

Price: £941.80
