

## Grandstream UCM6302 IP PBX

Product Name: Grandstream UCM6302 IP PBX

Manufacturer: Grandstream

Model Number: UCM6302

### Grandstream UCM6302 IP PBX

The Grandstream UCM6302 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.

### Grandstream UCM6302 Key Features

- Supports up to 1000 users and up to 150 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 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. The UCM6300 ecosystem consists of the Wave app for web and mobile, which provides a hub for collaborating remotely, and UCM Remote Connect, 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 UCM6302 - Technical Specifications

#### Analog Telephone FXS Ports

• 2 RJ11 Port

• All ports have lifeline capability in case of power outage

#### PSTN Line FXO Ports

• 2 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

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Yes (supports router mode and switch mode)

### Peripheral Ports

1\*USB 2.0

1\*USB 3.0

1\*SD card interface

### LED Indicators

None

### LCD Display

320x240 colour LCD with touch screen for Shortcut Keys and Scroll Bar

### 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, G722.1 G722.1C, G.723.1 5.3K/6.3K, G.726-32, G.729A/B, iLBC, GSM; T.38

### Video Codecs

H.264, H.263, H263+, 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

### DTMF Methods

In-band audio, RFC2833, and SIP INFO

## Grandstream UCM6302 IP PBX

### 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&reg;

### 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

• Input: 100 ~ 240VAC, 50/60Hz; Output: DC 12V, 1.5A

### Dimensions

• 270mm(L) x 175mm(W) x 36mm(H)

### Weight

• Unit Weight: 725g

• Package Weight: 1221g

### Temperature & Humidity

• Operating: 32 - 113&ordm;F / 0 ~ 45&ordm;C, Humidity 10 - 90% (non-condensing)

• Storage: 14 - 140&ordm;F / -10 ~ 60&ordm;C, Humidity 10 - 90% (non-condensing)

### Mounting

• Wall 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, Nederlands

• Customisable language pack to support any other languages

### Caller ID

• Bellcore/Telcordia, ETSI-FSK, ETSI-DTMF, SIN 227 &dash; BT, NTT

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### Polarity Reversal/Wink

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

### Call Center

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

### Customisable Auto Attendant

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

### Maximum Call Capacity

• Users: 1000

• Concurrent calls (G.711): 150

• Max concurrent SRTP calls (G.711): 100

### Maximum Attendees of Conference Bridges

• 3 Video Conference rooms and up to 20 parties with 1080p, assuming 4 video feeds + 1 screen sharing (H.264 & G.711)

• Voice Conference: Up to 150 parties (G.711)

### Wave Mobile App

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

### Call Features

• 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

• 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

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

• 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

**Grandstream UCM6302 IP PBX**

**Price: Â£322.40**

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