



Enterprise VolP and Softphone Conferencing, Extended

Larger conference rooms in the enterprise require additional microphones at the ends of the table to ensure quality call performance. Current solutions for this scenario involve connecting multiple units for additional speaker output and microphone input, which increases the overall system footprint for a minor improvement in audio quality.

The Revolabs FLXTM UC 1500 provides the answer for the larger conference room by building off the speakerphone capability of the FLX UC 1000 IP & USB Conference phone with two additional directional microphones.

The UC 1500's integrated SIP technology works in all major telephony environments, whether on premise or in the cloud, and supports all market-leading IP PBX's. In addition, the FLX UC 1500 easily integrates with your choice of 3rd party PC applications for collaboration and conferencing. A key benefit of this integration is that the FLX UC 1500 is capable of bridging VoIP Calls and calls via 3rd party softphone applications — allowing your team to communicate using the best medium for collaboration.

The FLX UC 1500 offers the most extensible solution for USB, VoIP, and Conference Bridging for Enterprise Conferencing and Collaboration.



FILLS THE CONFERENCE ROOM WITH HIGH QUALITY SPEAKER SOUND

In conference rooms sized for up to 18 people, speaker audio can be heard by every meeting participant; the FLX UC 1500's powerful speaker can fill the entire conference room with audio.

COMPLETE ROOM MICROPHONE COVERAGE

Every word by the meeting participants is picked up and transmitted to the far end due to the complete 360° coverage of the four built-in directional microphones and additional directional microphones.

FULL DUPLEX

Participants can communicate naturally as if they are in the same room, and the technology in the FLX UC 1500 supports simultaneous audio play back and audio capture ensuring every word is heard.

INTEGRATED ECHO CANCELLATION

The meeting participant speaking has his audio picked up and transmitted clearly; built-in digital signal processing separates out the speech of the meeting participant, room noise, and audio from the speaker of the device to ensure only the correct audio is transmitted to the far end. Real discussions between conference participants are finally possible.

INTUITIVE DIALER WITH FULL COLOR GRAPHICS DISPLAY

The keypad provides quick access to the key calling features: On-Hook, Off-Hook, Volume Up/Down, Mute, Redial, and Main Menu, with additional context-dependent soft keys, navigational buttons, and telephone keypad for ease of use and dialing. The intuitive dialer boasts an easy to use, icon-based menu system on the full color graphics display.

FEATURES AND SPECIFICATIONS

SIP

Call Handling

Call Bridging

Make Call, Answer, Hold, Resume, Transfer (Blind and Supervised), Forwarding, Do not disturb, Redial, Call Timer, Call ID, Speed Dial, Missed Call Notifications, Voice Mail Notifications (switch-configured) Supports Bridging SIP and USB Calls. Join, Split, Hold, Resume, 3+1 Lines: Up to 2 SIP calls, 1 USB call, plus the user

Codecs

G.711 (A-law and µ-law), G.722, G.726, G.729

DTMF Support Security

RTP event, SIP in-band, SIP info package support SRTP/SRTCP Support (RFC 1889), IETF SIP support (RFC 3261 and companion RFCs), SSIP, Password

protected configuration files, Password protected login, HTTPS Secure Provisioning

DHCP or Static IP, Server Based Central Provisioning, Web Portal for Individual Unit Configuration Network & Provisioning

DEVICE FEATURES

Base Unit Dialer Keypad

Dialer Display User Interface

Extension Mics

Capacitive Touch Buttons for On-Hook, Off-Hook, Volume Up/Down, Mute

12-key telephone keypad, 4 context-dependent soft keys, 5 Navigation buttons with select key, On-Hook,

Off-Hook, Volume Up/Down, Mute, Redial, Menu

3.5in diagonal color graphics display, 480x320 pixels

Date & Time Display, User-configurable Local Contact List, Call History (Missed Calls, Placed Calls, Received Calls), User-Selectable Ring Tones, Unicode UTF-16 Character Support, Multilingual User Interface Capacitive Touch Button for Mute and Status indicator

AUDIO FEATURES

Speaker Elements Speaker Frequency Response

MAX SPL Microphones

Microphone Coverage Microphone Frequency Response Echo Cancellation Full Duplex

Tweeter, Mid-Woofer 160 - 20,000 Hz 91 dB at 1m

4 embedded directional

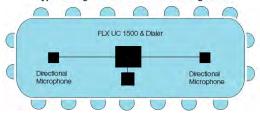
2 directional extension microphones 360° Audio Pick-Up

160 – 16.000 Hz

Individual AEC per microphone

Yes

Typical Large Conference Room Configuration



INTERFACES

Power USB Power over ethernet (PoE) Mini USB 2.0 Connector

ENVIRONMENTAL

Operating Temperature Storage Temperature Relative Humidity

32 - 104 °F (0 - 40 °C)

-22°F to +131 °F (-30°C to +55 °C)

20%-85% (noncondensing)

SYSTEM INCLUDES:

- 1 Base Unit
- 1 Dialer Unit
- 1 USB Cable, 6.56ft (2.0m)
- 1 Ethernet Cable, 25ft (7.62m)
- 2 Directional Extension Microphones
- Quick Start Guide

COMPLIANCE

Safety

CE Mark: EN 60950

IEC 60950

UL 60950

CSE C22.2, No. 60950

AS/NZS 60950

EMC

FCC (47 CRF Part 15) Class B

ICES-003 Class B EN55022 Class B CISPR22 Class B

AS/NZS CISPR22 Class B

VCCI Class B EN55024

Environmental

RoHS

Revolabs HQ

144 North Road Sudbury, MA 01776 USA +1 800-326-1088 sales_americas@revolabs.com

Revolabs EMEA

190 High Street Tonbridge, Kent TN9 1BE, UK



info@revolabs.com www.revolabs.com

SIZE, WEIGHT, POWER

Base Unit Height Base Unit Width Base Unit Length Dialer Unit Height Dialer Unit Width Dialer Unit Length Weight

Power

4.84 in (123mm) 5.9 in (150mm)

5.9 in (150mm)

1.6 in (41mm) 5.2 in (133mm)

5.3 in (136mm) 3 lb (1.3 kg)

IEEE 802.3af Power over Ethernet (PoE)

WARRANTY

1 year warranty *EU warranty is 2 years