UNIVERGE® 3C™ Media Gateways & Conferencing Bridge

At a Glance
- COHub
- PhoneHub
- BranchHub
- MeetingHub 60-Port conference bridge

Overview
UNIVERGE Media Gateways are easy-to-deploy network appliances. The COHub, BranchHub and PhoneHub provide PSTN connectivity and legacy PBX interconnection as well as IP-enabling analog stations for seamless migration to VoIP. The MeetingHub Conferencing Bridge delivers powerful on-demand conferencing for up to 60 callers. All of these gateways are integrally designed to work with the UNIVERGE 3C system and are automatically discovered by the call manager allowing for simple, quick deployment.

UNIVERGE Media Gateways and the MeetingHub Conferencing Bridge provide the tools that enable organizations to leverage their technology, protect their existing investments and fully meet employee and customer needs.

Solution

COHub
Digital Trunk Connectivity
- Connects UNIVERGE 3C Unified Communications Manager to the PSTN for outside calling via T1/E1 PRI, T1 CAS, E1/R2
- Connects UNIVERGE 3C Unified Communications Manager to an existing PBX via T1 Q.SIG to integrate branch offices or consolidate multiple PBXs
- Auto-discovery and configuration for easy deployment

BranchHub
Analog Station & Trunk Connectivity
- Connects UNIVERGE 3C Unified Communications Manager to the PSTN for outside calling via six analog loop start trunks
- Supports up to 12 analog stations per media gateway
- Integrates analog phones with UNIVERGE 3C UC Client software
- Power-loss transfer connects phones to the PSTN
- Auto-discovery and configuration for easy deployment

PhoneHub
Analog Station Connectivity
- VoIP enables analog stations including phones, fax machines, and more
- Integrates analog phones with UNIVERGE 3C UC Client software
- Supports up to 24 ports per media converter
- Power-loss transfer connects emergency phone to the PSTN
- Auto-discovery and configuration for easy deployment

MeetingHub
60 Port Conference Bridge
- In-house conference bridge enables up to 60 simultaneous participants across multiple conference lines
- Eliminates the need for expensive outside conferencing services
- Includes all software licensing at no additional charge
• Each conference bridge can be configured with multiple DIDs or extensions or can be deployed as private conference lines
• UNIVERGE 3C UC Client software displays the Caller ID of conference participants along with call duration
• Auto-discovery and configuration for easy deployment

Media Gateway & Conferencing Specifications

COHub Specifications
• Signaling protocols: T1/E1 ISDN, T1/E1 Q.SIG, T1 CAS, E1/R2
• Complies with AT&T TR62411 for clock and data recovery, jitter tolerance and other electrical standards
• Long Haul DS1 and Short Haul DSX1 digital network interfaces
• RJ48C jack for T1/E1 and two BNC jacks for E1 access
• Dual Bantam jack for T1/E1 monitoring
• Full touch-tone DTMF support

BranchHub Specifications
• 6 loop start FXO ports
• Ring detection: 20 Hz @ 40-130 VAC
• Caller ID detection per Telcordia GR-30-CORE
• 6 failover ports to FXS stations
• Software programmable for country-specific tone and timing requirements

Media Gateway Network Specifications
• Auto-discovery to UNIVERGE 3C IP PBX
• Auto-sensing 10/100 RJ45 Ethernet interface
• G.711, G.729 codec support
• Codec auto-negotiation
• A-law or μ-law companding
• Echo cancellation per G.168
• QoS: 802.1 p/q for Ethernet, DiffServ for IP
• DHCP or static IP addressing
• RTCP media, control, statistics and reporting
• TFTP for firmware updates
• Telnet
• SNMP

MeetingHub Specifications
• Up to 60 individual conference ports per Hub
• Multiple virtual conference bridge lines per Hub
• Conference call participants displayed and managed via Sphericall Desktop
• Automatic VoIP codec transcoding

Analog Station Specifications
• 24 FXS station ports
• 1 PSTN emergency failover port (PhoneHub)
• 6 PSTN emergency failover ports (BranchHub)
• 50-position telco connector, wired per RJ21X
• 25mA DC loop current (polarity insensitive, off-hook battery = -24VDC)
• 2 wire impedance, 600ohm nominal
• Ring generation: 20Hz, 2 secs on/4 secs off (default)
• 45Vrms, trapezoidal wave shape, 2.0A REN max
• Frequency response: 300 to 3400 Hz +/-3dB (tx/rx)
• Max loop length: 2000 ft. (609m), 24 AWG on-premises wiring
• 2500 analog phone set support
• CLASS phones and features including: Caller ID, Call Waiting with Caller ID, FSK Message Waiting, Park, Pickup, Do Not Disturb, Forwarding
• Paging system interface
• Music-on-hold source interfaces
• Modern and fax support
• Country specific and customizable tones and timings

Regulatory Compliance

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<th>Regulatory Compliance</th>
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<tbody>
<tr>
<td>FCC Part 15, Class A</td>
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<td>FCC Part 68</td>
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Functionality

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<tr>
<th>Functionality</th>
<th>Peripheral Support</th>
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<tr>
<td>Other</td>
<td>1U 19 inch (48cm) rack-mounted or standalone</td>
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<tr>
<td>Form Factor</td>
<td>100-240 VAC auto-sensing</td>
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<td>Normal Operating Temp.</td>
<td>32°F – 104°F, 0°C - 40°C</td>
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<tr>
<td>Normal Operating Temp.</td>
<td>0% to 90% RHNC</td>
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<tr>
<td>Dimensions / Weight</td>
<td>17 ⅞ x 1 ¾ x 15 in. (44 x 4.45 x 38.1cm), 10.8 lbs. (4kg)</td>
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