

OM80E IP-Telephony System



OM80E

Max. IP extensions	168	
Max. built-in analog extensions	48	
Max. IP trunks	216	
Max. built-in analog trunks	48	
Max. concurrent calls	90	
ВНСС	14.4K	
H×W×D (1U)	44×440×260 mm	

Product models

With modular design, the capacity of the OM80E can be flexibly expanded to 16-48 FXS/FXO ports.

Highlights

- Connected to IMS Networks via ONU
- Dual-redundant Ethernet ports
- Dual AC/DC power supplies (optional)
- Multi-site deployment supports up to 300 sites
- Call recordings on the internal storage or external server
- Add-on call recording management system
- Add-on middle-ware for hospitality applications
- XML/HTTP-based RESTful API interworking with and supporting third-party applications
- Multi-level auto-attendant
- Support TR-069, SNMP
- Interoperability with popular SIP servers, such as Cisco Unified CallManager (CUCM), Broadsoft, Microsoft Skype for Business (Lync), Huawei IMS, and Asterisk/Elastix
- Class I lightning protection

The Officium 80E (hereinafter "OM80E") is a highly integrated hybrid IP-PBX with up to 200 analog and/or IP extensions. Due to its high-performance and XML/HTTP based API, it is a cost effective solution for mid-size contact center with up to 60 agents as well.

Powered by innovative VoIP technology from New Rock, the OM80E provides seamless connectivity to not only PSTN and analog terminals, but also IP-based SIP trunks and SIP voice stations.

Always Connected with Your Customers

With features such as remote extension, call forking to your mobile phone, and call-back, the OM80E assures you that wherever you are--whether in your office, traveling abroad, or working at home--you will never miss any important business calls.

Improving Call-Answer Rate

The auto-attendant provides 24×7 service to customers, and efficiently directs incoming calls to the extensions or cell phones of the right person through configurable voice menus in business/non-business hours and holidays. The smart auto-attendant can even deliver the calls from VIP or repeat customers to the appropriate agents directly, bypassing voice menus (caller-agent binding).

Cost Reductions

The OM80E has all of the IP remote-calling functions needed to support the remote worker, business travel, and communications with branches, all through the Internet. International or long-distance calls are turned into local toll-free calls. With advanced routing capabilities, the OM80E can route calls intelligently according to call types. Combined with New Rock's WeWei [®] softphone app and NeeHauTM business-phone assistant, you can communicate with your customers effectively, regardless of where you are.

Remote Management and NAT Traversal

The OM80E includes an embedded New Rock Cloud client that allows the OM80E located behind an enterprise NAT or firewall to be accessed from the Internet for maintenance and troubleshooting or by voice devices from remote branch sites, remote IP extensions, or far-end value-adding applications. Real-time monitoring, alarm notification, remote packet capture and software upgrades can be performed with the New Rock Equipment Management System or third-party device management systems with TR-069 or SNMP.



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Specification

Protocols

Call control SIP/UDP and SIP/TCP (RFC3261), IMS (3GPP)

Network Telnet, SSH, HTTP, HTTPS, DHCP/PPPoE client,

DNS (A/SRV record), STUN

Media Processing

Caller ID Bellcore Type 1&2, ETSI, BT, and DTMF-based CID

Codec G.711 (a/µ), G.729a

DTMF In-band audio, RFC2833, SIP-INFO

Hook-flash Local processing, RFC2833, SIP-INFO

Fax over IP T.38, G.711 pass-through

T.38 compliant Group 3 Fax Relay

Maximum fax rate of 33,600 bps (pass-through)

Disconnect modes Polarity reversal, Busy tone detection, Loop current

Voice-quality Echo cancellation (G.168-2004), Jitter buffer,

enhancement Silence suppression (VAD, CNG), PLC

Voice

Smart autoattendant/ Receptionist

Business/non-business hours/holiday, Music on hold, Attendant group, Multilingual/multi-level IVR,

Auto attendant profiles, VIP service

Dialing Speed dial, Outgoing route selection, Least-cost

routing, Automatic route selection

Call settings Call transfer, Call forward, Do not disturb, Barge,

Silent monitoring, Simplified DISA etc.

Call recording Through internal storage or New Rock pbx record

software

Voicemail MWI (polarity reversed, FSK, high voltage)

Analog trunk Caller ID, Polarity reversal detection

SIP trunk IMS, Multiple SIP servers, Skype Connect

Phone status Status subscription (BLF)

indication

Provisioning, Administration and Maintenance

Device New Rock Equipment Management System,

management TR-069 management (TR-069, TR-104, and

TR-106), SNMP

Remote access over TCP

New Rock Cloud

Auto provisioning Download configuration file via TFTP/FTP/HTTP/

HTTPS, Obtaining ACS address via DHCP option

66 or redirection

API XML/HTTP

Log 8-level log management, Syslog, Internal USB

flash drive (16 GB)

Data capture Port capture, Packet capture

 Configuration
 Bulk import/export. Resetting to the factory default

 Status and
 Call status and history, Device status monitoring and

statistics

Odii status and history, Device status monitoring an

statistics collection

Monitoring/ alarming

Status/performance monitoring or reporting.

Status/performance alarms

Other

Multi-site voice Multi-site networking across NAT without the need

of static IP address, DDNS or port mapping on

the access router.

Uniform dial plan, Call forward, Call transfer.

Outbound-trunk sharing, Three-way calling,

Attendant on remote site

QoS DiffServ, TOS, 802.1p/q VLAN tagging

Internal storage Local recording, Voicemail, Customer information,

User-uploaded IVR file

Security

User-defined SIP port, RTP port, HTTP/HTTPS port to access ports

the Web GUI

Access list IP addresses allowed to access HTTP/HTTPS/

Telnet/SSH service, IP address filtering of SIP

IP phone protection

Prohibition on outgoing dialing by IP extensions

on public network, User-Agent authentication, Registration password cracking protection

Web GUI login protection

Prohibition on login from public IP address, Login

password cracking protection

Encryption Encryption on SIP signaling or/and media streams.

Importing and exporting encrypted configuration

file and password/PIN

Security level Three levels of security settings

Hardware

CPU	1 GHz	RAM	256 MB, DDR3	
Flash	32 MB	Internal storage	16 GB	
Mounting	Rack	Net weight	3 kg maximum	
Voice DSP module	200 MHz, 2 to 6 modules			
Single/Dual AC power supplies	100 to 240 VAC, 50/60 Hz, 1.2A maximum			
Single/Dual DC power supplies	-36 to -72 VDC, 2.5A			
Power consumption	60 W maximum			
Operating	Temperature: 0 to 40°C, Humidity: 10% to 90% RH (non-condensing)			
Storage	Temperature: -40 to 70°C, Humidity: 5% to 90% RH (non-condensing)			

