

THE FLEXIBLE COMMUNICATIONS SOLUTION FOR SMALL AND MEDIUM BUSINESS



OPERA 240 SME VOIP Switch



THE COMPLETE BUSINESS SOLUTION FOR INTERNET TELEPHONY

Opera 240 integrates all the voice communications needs of a small to medium business on a modular IP platform which can be configured for SIP trunks and ISDN lines, with integrated voicemail, automated attendant, FMC and CTI.

Operafone IP Professional



Operafone IP Executive with DSS Module



Operafone Soft for Windows PC



Operafone Soft for iPhone

Configuration example showing an Opera 240 rack with plug-in cards. The rack has 8 FXS analogue POTs as standard. This rack has a plug-in card for 8 Upn digital system phones and two PRI cards.

Modular system

The basic 19" rack accommodates up to 240 IP users. 8 FXS analogue a/b interfaces are standard. Up to 60 SIP trunks may be provisioned.

Plug-in cards

The basic unit has three universal plug-in slots into which any three of the following cards may be inserted.

ISDN PRI card (max 2 cards), each with 1 PRI ISDN BRA card, with 4 x T0 basic accesses. FXS Analogue POTs extensions' card, with 8 a/b. Upn Digital extension card, with 8 system phone interfaces.

Expansion racks

Additional expansion racks, each with three slots for plug-in cards, may be connected to the LAN, to increase the hardware capacity, as per following examples, per expansion rack added.

32 FXS analogue a/b POTs users

24 Upn digital system phones and 8 FXS Up to 240 TDM extensions, between FXS and digital, can be connected using expansion racks as Slaves off the Master rack.

The Slaves are tightly coupled with the Master, giving presence information across all registered users. Access to outside lines, either ISDN or SIP trunks, is exclusively via the Master.

IP System phones

IP keysets may be connected locally on the LAN or remotely over Internet. Remote IP extensions have full keyset functionality. A PC may be plugged into an Ethernet mini-hub on the back of the IP system phone, allowing a single cable to the desk.

There are two desktop IP system phones to choose from, Executive and Professional, each with a 4 line backlit display, with context sensitive navigation keys. A 48 key Direct Station Select (DSS) module may be used in conjunction with the Executive key set.

The Executive set is also available as a PC softphone application, offering keyset functionality on any PC running Microsoft Windows. It's also available as an iPhone app. An Android app is in development.

Digital System phones

2-wire digital Executive keysets may be connected at the Upn digital plug-in card

Email forwarding of voice mails and recorded calls

Individual and group voice mails or recorded calls may be forwarded automatically to your email address as a WAV attachment. You can alter your voicemail greeting remotely and change your call forward destination from your mobile phone.



Automated Attendant

One hundred separate interactive messages and menus can be played to callers depending on the number they dialed, the time of day, the language preference and the other options selected by the caller.

VoIP Networking of multiple systems

Up to one hundred Opera systems can be networked over Internet using two or more IP channels per system. A single numbering plan covers all of the networked systems. A common operator may be programmed to ring for incoming calls, across any systems.

Browser-based user portal

Download your system phone book from Gmail or Outlook. Set Call Forward and manage voice mails and general user parameters.



Dial from your PC using CTI

The desktop PC can dial calls from Outlook. Incoming calls can pop a screen on the PC, providing access to the relevant customer database information.

Fixed Mobile Convergence (FMC)

A user may have single number reachability and a single, centralised mailbox for the desk phone, home system phone, mobile and soft phone.

Any WiFi/3G and SIP compatible mobile may be registered to a user as an extension device on the wireless LAN or roaming on the 3G public network. The incoming caller ID of the originating caller is presented to such handsets.

The user presents one CLI, irrespective of the device used to make the call.

For a user with several registered devices (home, office, mobile, soft), the last device used is the active device, to which all incoming calls are directed. The user can change active devices by either going off line, for example in the case of a soft phone, or by pressing a key to push activity to another device or by pulling activity to another device by using it to make a call.

Feature/expansion cards & licences

SIP or ISDN trunks (to max 60 trunks in total) IP system phone users on the LAN (max 240). Digital or analogue a/b extensions (to max 240 users in total)

Multi-layer Automated Attendant

Voicemail boxes (users/groups: max 300 boxes) Call Recording

Email forwarding of voice mails and recorded calls Secure (AES) remote IP extensions over Internet Music-on-Hold, downloadable 3 min wav/MP3 CTI

PC Soft phone iPhone system phone app Networking of multiple Opera systems Meet-me conference rooms SIP extensions on the LAN 3G SIP mobile handset extensions

Automated Management Server

Zero Touch provisioning and configuration management. Order entry. Remote download of feature licences. SIP trunk provisioning. Firmware upgrades. Reboot. Backup/restore full system configuration. View system status, connected etc. Automated remote diagnostics. Service delivery without site visits.

Auto discovery by IP system phones

Set-up programming is not required for IP system phones. Just plug into the LAN, which has a DHCP server, and the phones auto discover the system and are automatically assigned extension numbers.

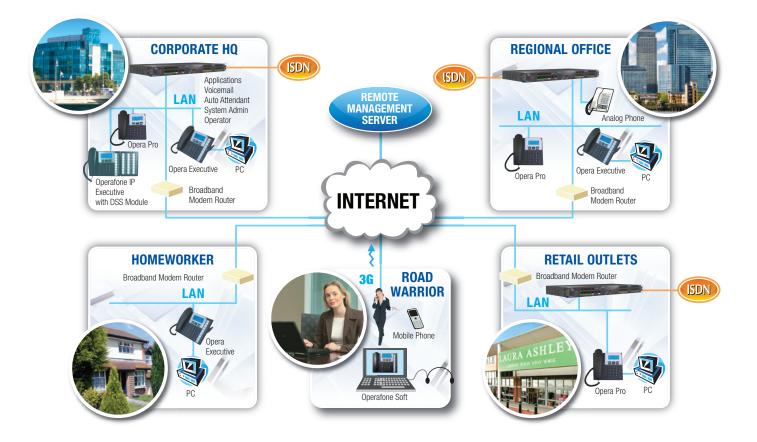
For home workers, a system phone that has auto discovered on a company LAN where the broadband modem/router supports Universal Plug n Play, can be plugged into any LAN with Internet access and it will automatically go on line and connect to its system as a remote extension.

Call Recording

Calls can be recorded and forwarded by email, as wav attachments.

Meet Me Conference Rooms

Two seven-party dial-in audio conferences may be held at the same time and can be recorded and emailed to participants.



OPERA 240 SME VolP Switch



TECHNICAL OVERVIEW

Switch capacity

Up to 60 SIP trunks, RFC3261 1/2 PRI, 4/8/12 BRA trunks Up to 240 IP users 8 analogue a/b extensions (FXS), expandable to 240 Up to 240 digital system phones Ethernet port External relays x 2

IP system phones

Four-line backlit display Two Ethernet ports (mini-hub) DC jack for 12 Volt plug-top adaptor or PoE Headset socket (Executive version only) DSS module (Executive version only)

Digital system phones

2 - wire Executive keyset Four-line backlit display Headset socket DSS module

Voice over IP

Voice compression codecs: G.711, G.729 Quality of Service (QoS): 802.1p/q Echo and silence suppression Echo cancellation G.165/G.168

Security

Remote IP keysets: AES encryption

System management

Browser based local or remote programming LEDs for power and status Local or remote configuration backup Local or remote software upgrade Remote diagnostics

Automatic configuration server

Automated Zero Touch provisioning with: Installation of optional licences Firmware upgrades Reboot Backup/restore of full system configuration View system status, firmware revision, Reset to factory default Automated remote diagnostics

Physical interfaces

Trunks: ISDN PRI, TO RJ45 Ethernet RJ45 a/b extensions (FXS) RJ45 Digital keysets RJ45 External relays (x 2) RJ45 Optional audio adaptor for public address, at an FXS port, RJ45

Power

110V / 220V / 240V ac (market dependent) Switch consumption: typical 24VA Plug-top adaptors for IP keysets PoE IEEE 802.3 af

Status LEDs

Power LAN Internet, SIP server Error warning Expansion card 1 Expansion card 2 Expansion card 3 Expansion rack (Slave), Exp

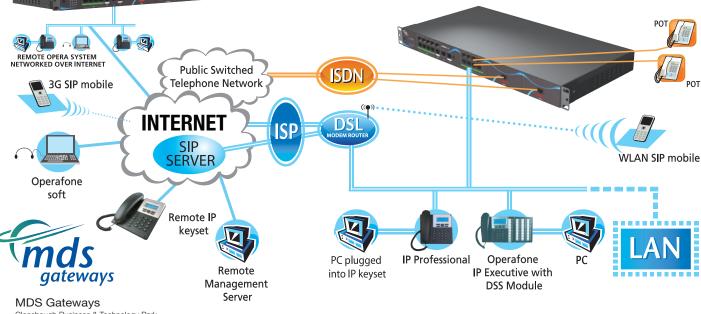
Physical dimensions

19" rack full width, one height unit, three quarter depth Weight: 2Kg Rack and wall mountable

Voice functionality

Alarm clock calls Analogue CLI Auto-discovery of IP stations Automated attendant Automated Management Server Automatic Call Distributor Browser based system programming Browser based user programming Call-back Call Barring, up to 7 levels Call Forward Forward all Forward no answer Forward on busy External or internal divert Reprogramme diversions remotely Call Hold Call Lists Missed Dialed Received Callback Call Logging Call Park Call Pick-up/ Call Pick-off Call Record Call Protection Call Transfer Call Waiting (internal and external) CLIP, CLIR Conference Call (Internal and External) CSTA CTI Do not disturb Doorphone (1 x a/b extn, 2 x relays) Dynamic DNS Email forwarding of voice mails and recorded calls Embedded web server Enquiry Call External line ringing assignment Fixed Mobile Convergence (FMC)

Gmail contacts' synchronisation Groups: Dedicated mailboxes Group diversion Group mailbox Missed call list per group Up to 30 groups Headset ready Hot desking Hotline Hunting groups IP server (for IP extensions) IP system phones Outlook contacts' synchronisation Least cost routing Meet-me conference rooms MSN ringing assignment MSN restriction Multiple user devices (desk, soft, mobile) Music on hold Networking of multiple systems Night service Paging via system phones Phone Book pool of 3,500 entries import / export of directories Internal extension list Programmable keys 16 Re-dial (last 10 numbers) Ringing Modes, 5 SIP client (up to 60 SIP trunks) SIP extension users Soft phone Synchronisation with Outlook. Gmail contacts iPhone app UPnP (Universal Plug and Play) User PIN codes User profiles (Office/Home/Roaming) Voicemail Voice recording WLAN/3G SIP mobiles



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