

RADCOM ENSURES A WORLD OF IP SERVICES IN YOUR NETWORK

RADCOM, an acknowledged leader in network test and quality management solutions, offers innovative multi-technology testing equipment for VoIP, Cellular and Datacom system performance, providing session, network and service level analysis for troubleshooting and monitoring the IP, UMTS, GPRS and CDMA2000 networks, as well as a range of applications for establishing, maintaining and monitoring VoIP networks.

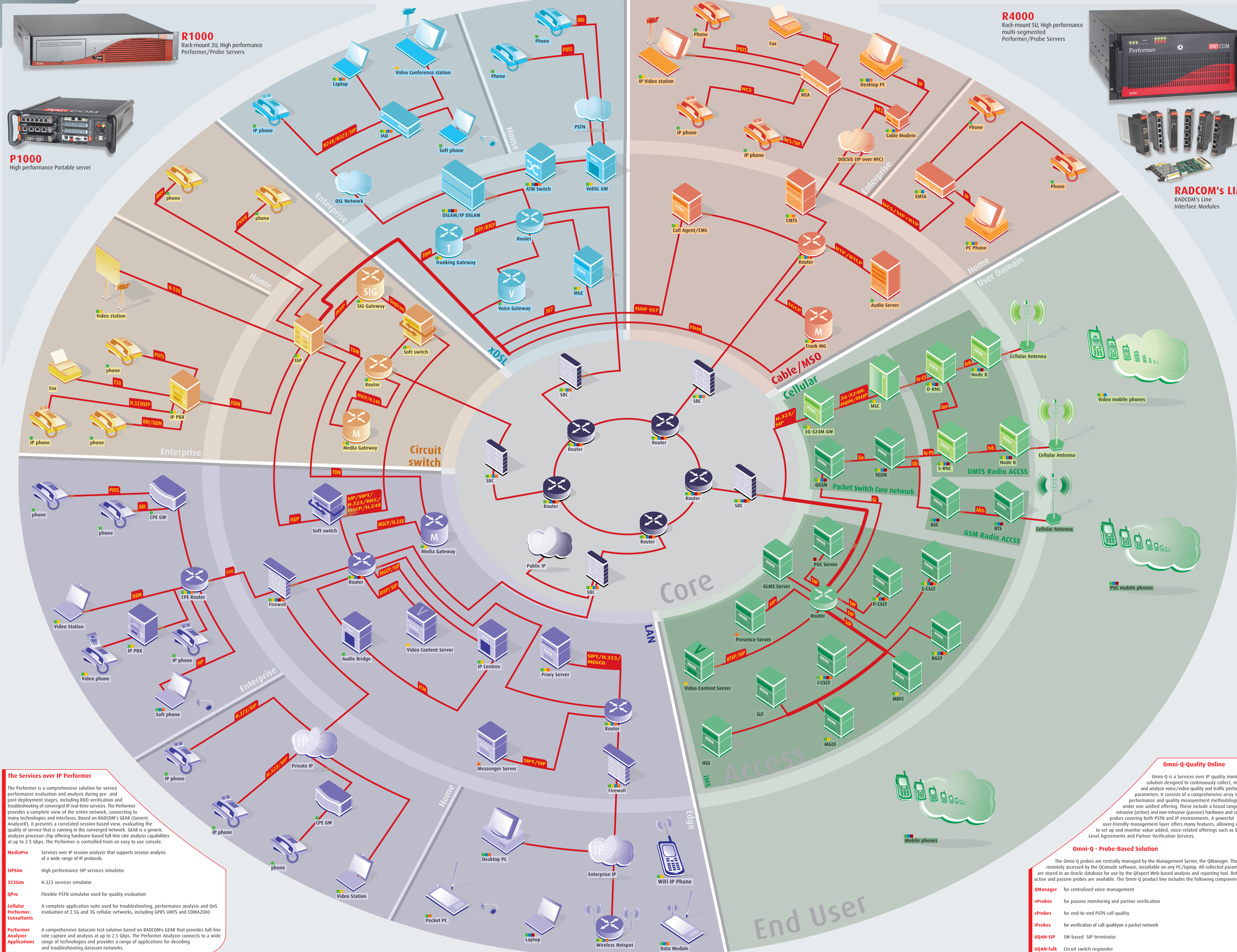
TEST-OF-THE-ART

Services

- Centrex** Central Office Exchange service in which switching occurs in the local telephone station instead of at company premises
- Data Service** Ability to access services which are not time sensitive, such as Web browsing, File transfer, Mail
- IVR** Interactive Voice Response, an interactive voice application that accepts touch-tone telephone keypad and provides appropriate response (such as voice, fax and call-back call)
- Presence & IM** Presence is a service that defines the status of a user PC, IP phone or PDA in the network. The Presence service allows the configuration of whether, and by whom to be reached
- IM (Instant Messaging)** is the ability to send short messages in near real time to other IM users who are connected to the IM server, according to its status
- PTT** Push To Talk over services, a half-duplex form of communication that allows users to engage in immediate communication with one or more receivers. PTT services are used in the 2.5G & 3G cellular networks known as POC (Push to Talk over Cellular). Working group: 3GPP, 3GPP2, IETF. Additional developed PTT-based services are PIV (Push to View).
- SMS/MMS** SMS, Short Message Service, similar to paging service for sending text messages between phones
MMS, Multimedia Message Service, a method for transmitting graphics, video clips, etc.
- Video** Enables motion pictures in addition to audio communication through the network of two participants or more. Several common video services are:
Video streaming (Video On Demand) Live playback of video files through the network. Can be done by file download (clips), or real-time transmission using a buffer with no need for local storage. There are two types of video streaming applications: playback service (movie playback) and real-time streaming such as IPTV.
Video telephony Enables video and audio conversation through the network of two participants or more.
Interactive video Video streaming service that can be changed during the session according to user action (such as clicking a button or digit).
Gaming Often used as video game. Service that offers interactive games played on a video screen with one or more participants in different locations.
- Voice** Enables audio conversation through the network of two participants or more
- Voice/Video Mail** Email system that supports voice/video. Ability to open by mail system voice and/or video messages

Technologies

- ATM** Asynchronous Transfer Mode used for high data transfer
- CDMA** Code Division Multiple Access Method for describing physical radio channels. CDMA2000 is CDMA technology that meets 3G requirements
- DSL** Digital Subscriber Line technology that allows data transmission of up to 6.1 Mbps using ordinary telephone lines
- Ethernet** Baseband LAN used to connect computers, workstations, terminals and other devices
- GPRS** Offers data throughput in the 2.5G of up to 300Kbps; connects mobile cellular users to a public data network (PDN) within a GSM network
- HFC** Hybrid Fiber Coax network that contains both fiberoptic cables and copper coaxial cables to carry TV signals from the head-end office to the neighborhood
- IP** IP Internet Protocol (layer 3 protocol) that includes services such as address, fragmentation & reassembly, security
TCP Transmission Control Protocol; reliable octet streaming protocol
UDP User Datagram Protocol; connectionless protocol mainly used for real-time services
- MBWA** Mobile Broadband Wireless Access IEEE 802.20 working group aims to prepare formal specification for packet-based air interface designed for IP services
- UMTS** Universal Mobile Telecommunications System
- WCDMA** Wideband Code Division Multiple Access derived from CDMA and meets 3G requirements
- WiFi** Wireless Fidelity standard of Wireless Local Area Network based on the IEEE 802.11 specification
- WiMax** Broadband wireless access technology based on the IEEE 802.16 working group, considered Metropolitan Area Network technology



The Services over IP Performer

The Performer is a comprehensive solution for service performance evaluation and analysis during pre- and post-deployment stages, including R80 verification and troubleshooting of converged IP real-time services. The Performer provides a complete view of the entire network, connecting to many technologies and interfaces. Based on RADCOM's GEAR (Generic Analysis), it presents a correlated session-based view, evaluating the quality of service that is running in the converged network. GEAR is a generic analysis processor chip offering hardware-based full-line rate analysis capabilities at up to 2.5 Gbps. The Performer is controlled from an easy-to-use console.

- MediaPro** Services over IP session analyzer that supports session analysis of a wide range of IP protocols
- SIPsim** High performance SIP services simulator
- 3235sim** H.323 services simulator
- QPro** Flexible PSTN simulator used for quality evaluation
- Cellular Performer Consultants** A complete application suite used for troubleshooting, performance analysis and QoS evaluation of 2.5G and 3G cellular networks, including GPRS UMTS and CDMA2000
- Performer Analyzer Applications** A comprehensive datacom test solution based on RADCOM's GEAR that provides full-line rate capture and analysis at up to 2.5 Gbps. The Performer Analyzer connects to a wide range of technologies and provides a range of applications for decoding and troubleshooting datacom networks.



R4000
Rack-mount SU, High performance multi-segmented Performer/Probe Servers



P1000
High performance Portable server



RADCOM's LIMS
RADCOM's Line Interface Modules

Legend

- Voice
- Video
- Data
- Poc
- Presence & IM

Protocols	Description
3G-324M	Standard that represents real-time streaming of wireless multimedia services over existing circuit-switched wireless networks
BICC	Bearer Independent Call Control implementation of SS7, a recommendation of ITU-T Q.850 or ANSI T1.602
BRI	Basic Rate ISDN
DOCSIS	Data Over Cable Systems Interface Specification listing of standards for modems and related equipment for data communications via cable television systems
H.320	ITU standard for ISDN conferencing of video, audio and data
H.323	IP protocol that is extension of ITU-T H.320 that enables multimedia
ISDN	Integrated Services Digital Network that allows high-speed access to the Internet using the telephone network
ISUP	ISDN User Part used to manage trunk circuits that carry voice and data over PSTN
Loopstart	Signaling used by analog phone lines which allows indication of onhook / offhook and ring / no-ring
Megaco	Media Gateway Control
MGC	Media Gateway Control Protocol
NCS	Network-based Call Signaling
POTS	Plain Old Telephone Service
PRI	Primary Rate ISDN
RTSP	Real-Time Transfer Protocol used in IP networks to provide end-to-end transport of delay-sensitive multimedia traffic
RTP	Real-Time Control (Protocol) protocol that monitors the QoS of the RTP connection and reports on the on-going quality of the RTP stream
RTP XR	Real-Time Control Protocol Extended Report) conveys information assessing the IP call quality and diagnosis problems
RTSP	Real-Time Streaming Protocol
SGCP	Simple Gateway Control Protocol
Signalt	Signaling Transport used to bridge between SS7 signaling and VoIP networks
SIP	Session Initiated Protocol
SIP-I	SIP for telephones a mechanism that facilitates the interconnection of the PSTN with SIP-based networks
T.30	ITU recommendation for document facsimile transmission in the general switched telephone network
T.38	Standard that represents transmission of fax over IP
TGCP	Trunking Gateway Control Protocol

Glossary	Description
MGC	Media Gateway Control
BSC	Base Station Controller
BTS	Base Transceiver Station
CMS	Call Management System
CMTS	Cable Modem Termination System
CPE	Customer Premises Equipment any terminating hardware owned by the user and not by the telephone company
D-RNC	Drift Radio Network Controller
DSLAM	DSL Access Multiplexer
EMTA	Embedded Multimedia Terminal Adapter
FTTC	Fiber To the Curb
FTTH	Fiber To the Home
FTTN	Fiber To the Node or Neighborhood
GSN	Gateway GPRS Support Node
GLMS	Group and List Management Server
HFC	Hybrid Fiber-Coax
HSS	Home Subscriber Server
HSS	Home Subscriber Server
IAD	Integrated Access Device
I-CSCF	Interrogating Call Session Control Function
IMS	IP Multimedia Subsystem IP Multimedia and telephony core network defined in the 3GPP and 3GPP2 standards
MGC	Media Gateway Controller
MGC	Media Gateway Controller
MRFC	Media Resource Function Controller
MRFP	Media Resource Function Processor
MSC	Mobile Switching Center
MTA	Multimedia Terminal Adaptor
Node B	UMTS Base Station
P-CSCF	Proxy Call Session Control Function
POC	Push-to-Talk Over Cellular
PSDN	Public Switched Telephone Network
SBC	Session Border Controller new layer 3 network element that provides carrier grade, secure, protocol-aware session-based network translation, firewall traversal, and QoS/SOR functionality for real-time, multimedia communications
S-CSCF	Serving Call Session Control Function
SGSN	Serving GPRS Support Node
SLF	Subscriber Location Function used in large IMS networks where there are several HSS entities
S-RNC	Serving Radio Network Controller
UMA	Unlicensed Mobile Access network between WiFi and GSM / GPRS networks

Omni-Q-Quality Online

Omni-Q is a Services over IP quality monitoring solution designed to continuously collect, monitor and analyze voice/video quality and traffic performance parameters. It consists of a comprehensive array of voice performance and quality measurement methodologies, all under one unified offering. These include a broad range of intrusive (active) and non-intrusive (passive) hardware and software probes covering both PSTN and IP environments. A powerful user friendly management layer offers many features, allowing users to set up and monitor value-added, voice-related offerings such as Service Level Agreements and Partner Verification Services.

Omni-Q - Probe-Based Solution

The Omni-Q probes are centrally managed by the Management Server, the QManager. They are remotely accessed by the QConsole software, installable on any PC/Laptop. All collected parameters are stored in an Oracle database for use by the QExpert Web-based analysis and reporting tool. Both active and passive probes are available. The Omni-Q product line includes the following components:

- QManager** for centralized voice management
- vProbes** for passive monitoring and partner verification
- eProbes** for end-to-end PSTN call quality
- iProbes** for verification of call quality on a packet network
- UQAN-SIP** SW-based SIP terminator
- UQAN-talk** Circuit switch responder