

Temas de Monografía 2009

1- Broadband over powerlines, BPL Broadband over power lines (BPL), also known as power-line internet or Powerband, is the use of PLC technology to provide broadband <u>Internet access</u> through ordinary power lines. A computer (or any other device) would need only to plug a BPL "<u>modem</u>" into any outlet in an equipped building to have high-speed Internet access. Coordinador: Alejandro Dantiacq Alumno: Bertagna, Pablo

2- IEEE 802.15.4a

Specifies two additional PHYs using <u>Ultra-wideband</u> (UWB) and <u>Chirp Spread Spectrum</u> (CSS). The UWB PHY is designated frequencies in three ranges: below 1 GHz, between 3 and 5 GHz, and between 6 and 10 GHz. The CSS PHY is designated to the 2450 MHz ISM band.^I

Coordinador: Jorge Robles

Alumno:

3- Voice over IP Security Ver SIP Digest, SIPs, SRTP, H.235, ZRTP, SPIT, TLS Coordinador: Armando Lima/Patricia Clérigo

Alumno:

4- MOME IP monitoring and measurement http://www.ist-mome.org/about/ Coordinador: Santiago Pérez

Alumno:

5- IEEE 802.11e

IEEE 802.11e as of late 2005 has been approved as a <u>standard</u> that defines a set of <u>Quality of Service</u> enhancements for <u>LAN</u> applications, in particular the <u>802.11</u> <u>WiFi</u> standard. The standard is considered of critical importance for delay-sensitive applications, such as Voice over Wireless IP and <u>Streaming Multimedia</u>. The protocol enhances the <u>IEEE 802.11</u> <u>Media Access Control</u> (MAC) layer. Coordinador: Gustavo Mercado

Alumno:

6- Sensor Web Enablement WG

OGC members are specifying interoperability interfaces and metadata encodings that enable real time integration of heterogeneous sensor webs into the information infrastructure. Developers will use these specifications in creating applications, platforms, and products involving Web-connected devices such as flood gauges, air pollution monitors, stress gauges on bridges, mobile heart monitors, Webcams, and robots as well as space and airborne earth imaging devices. Coordinador: Cristian Pérez

Alumno:

7- Anycast en IPv6

Anycast is a network addressing and routing scheme whereby data is routed to the "nearest" or "best" destination as viewed by the routing

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Alumno:

8- Path MTU Discovery

The Internet Protocol defines the "path MTU" of an Internet transmission path as the smallest MTU of any of the IP <u>hops</u> of the "path" from the source address to the destination address. Or, looked at another way, the "path MTU" defines the largest MTU value that can traverse the "path" without the packets sufferring further fragmentation Coordinador: Alejandro Dantiacq

9- ENUM

Telephone Number Mapping (ENUM or Enum, from TElephone NUmber Mapping) is a suite of protocols to unify the telephone numbering system E.164 with the Internet addressing system DNS by using an indirect lookup method, to obtain NAPTR records. The records are stored at a DNS database. Coordinador: Gerardo Gosetto

Alumno:

10- TTCN-3 (Testing and Test Control Notation version 3) is a strongly typed test scripting language used in conformance testing of communicating systems and a specification of test infrastructure interfaces that glue abstract test scripts with concrete communication environments. TTCN-3 has been developed by ETSI and its predecessor is TTCN-2. Despite sharing same fundamental concepts, TTCN-2 and TTCN-3 are essentially two different languages, the latter having simpler syntax and standardized adapter interfaces. TTCN-3 has been used to test SIP, WiMAX, and DSRC test systems. TTCN-3 scripts can be combined with ASN.1 type definitions. ASN.1 is natively supported by major TTCN-3 tool vendors. Coordinador: Santiago Pérez

Alumno:

 SILK - nuevo codec utilizado por Skype y liberado (free) a la competencia, no OPENSOURCE. Especificaciones y Características técnicas. Coordinador: Patricia Clérigo

Alumno:

12- VoIP Codecs

Análisis de los codec de audio y video existentes. Especificaciones y Características Técnicas. Comparativa. Coordinador: Patricia Clérigo

Alumno:

13- Internet Protocol Television (IPTV) se ha convertido en la denominación más común para los sistemas de distribución por subscripción de señales de <u>televisión</u> y/ó <u>vídeo</u> usando conexiones de <u>banda ancha</u> sobre el <u>protocolo IP</u> <u>http://es.wikipedia.org/wiki/IPTV</u> Coordinador: Gustavo Mercado

Alumno:

14- IP ECN (Explicit Congestion Notification)

Network congestion avoidance is a process used in <u>computer networks</u> to avoid <u>congestion</u>. The fundamental problem is that all network resources are limited, including <u>router</u> processing time and link <u>throughput</u>. Another approach is to use IP ECN. ECN is only used when the two hosts signal that they want to use it. With this method, an ECN bit is used to signal that there is explicit congestion Pegado de <<u>http://en.wikipedia.org/wiki/Network_congestion_avoidance></u>

Coordinador: Alejandro Dantiacq

Alumno:

15- VoIP Traffic generators Trade-off

- <u>Codima autoVoIP Traffic Simulator</u> Simulate and Stress Test all types of VoIP networks
- <u>Spirent Communications</u> Test Solutions for VoIP networks and devices
- <u>Candela Technologies</u> LANForge FIRE VOIP/RTP/PESQ call generator
- <u>Empirix</u> Signaling and Media load and feature testing
- GL Communications
 - PacketGen generates SIP calls with or without RTP traffic
 - PacketScan monitor, collect, and analyze QoS statistics on VOIP traffic
- <u>Integrated Research</u> Prognosis will simulate, record and analyze VOIP traffic in real time.
- <u>lperf</u> creates network traffic and measures performance
 - Can be used to test a network to see how it might perform with increased VOIP traffic
- Ixia VOIP traffic generators and Network assessment tools
- <u>MyVoIPSpeed</u> simulates VoIP traffic over your Internet connection, measures key diagnostics including Jitter and Packet Loss, and provides an analysis of the voice quality
- <u>PacketIsland</u> 4"x4" in-line micro-appliances used in a distributed multi-site enterprise or SME to generate live VoIP traffic and measure loss, jitter, MOS, route performance, route flaps, etc. Also measures ongoing data traffic in network.
- <u>Sipp</u> SIP Performance Test Tool Performance tester for SIP
- <u>pjsip-perf</u> Open source call generator from <u>pjsip.org</u> to measure SIP call/transaction performance.
- <u>Touchstone</u> 100% software-based VoIP and video verification tools.
 - WinSIP SIP signaling and Audio/Video media generator
 - Win323 H.323 signaling and Audio/Video media generator
- <u>Valid8.com</u> Valid8.com is a leading provider of SIP, H.323, Megaco, SIGTRAN traffic generation solutions.

Alumno:

16- VoIP Monitoring and Test Tools Trade-off

- <u>ACQUA</u>: VoIP Speech Quality Analysis System
- <u>Agilent Technologies</u> DNA MX and TNA software
- <u>Brix Networks</u> real-time VoIP performance management and service assurance solutions
- ClarifiedNetworks Monitoring and high level flow analysis and visualization
- <u>ClearSight Networks</u> ClearSite Analyzer
- <u>Codima Toolbox</u> VoIP Management from pre deployment to post deployment and network visualization in Visio
- <u>Consultronics</u>
- <u>Empirix</u> monitoring and analysis
- Fluke Networks OptiView VoIP, ProtocolExpert Plus and Link Analyzer
- Hammer Call Analyzer
- Inet GeoProbe IP
- Intelica VoIP Intelligence Call Monitoring, Analysis and Reporting
- <u>Malden</u>
- <u>Minacom QOS</u> monitoring and testing
- <u>NetTest</u> monitoring of <u>OOS</u> and network performance
- <u>NetIO</u> monitoring of <u>OOS</u> and network performance
- <u>PacketIsland</u> micro-appliances and asterisk agent for QoS monitoring of SIP and Cisco SCCP VoIP calls
- <u>PathSolutions</u> SwitchMonitor VoIP
- Packet Data Systems Clarinet Protocol Test System
 - Simulation and analysis of <u>SIP</u>, SIP-T & <u>H.323</u>. <u>QOS</u> measurement, recording, replay etc.
- <u>Psytechnics</u> monitoring of <u>QOS</u>
- <u>Oovia</u> monitoring of <u>OOS</u>
- <u>Rochelle</u> Analog QOS products can be used with an <u>ATA</u>
- <u>Sage Instruments</u> VOIP test equipment and systems
- Spirent Communications Test Solutions for VoIP networks and devices
- <u>Sunrise Telecom</u> VOIP and Telecom test equipment

- <u>TamoSoft CommView</u> and <u>CommView for WiFi</u> network analyzers for wired and wireless (802.11 a/b/g/n) networks
 - Real-time VoIP call monitoring
 - SIP and H.323 analysis and decoding, call playback
 - Jitter, QoS, Bandwidth charts
- <u>Telchemy</u> monitoring of <u>QOS</u>
 - Their technology appears in numerous other products list
- <u>Touchstone</u>
 - WinEyeQ
 - 100% software-based
 - monitors/analyzes/records/replays <u>SIP</u> and <u>H.323</u> traffic, audio/video media and <u>QOS</u>.
 - TraceBuster
 - records/replays <u>SIP</u> and <u>H.323</u> traffic, audio/video media and <u>OOS</u>.
- <u>Unsniff Network Analyzer</u> RTP/SIP/IAX2 visual analysis, one-click playback, whitepapers
- <u>Valid8.com</u> Valid8.com is the leading provider of SIP, H.323, Megaco, SIGTRAN conformance test solutions.
- <u>Viola Networks</u> NetAlly RealTime and VoIP Assessment
- <u>VQManager</u> web-based, 24 X 7 real-time QoS monitoring tool for VoIP networks.
- <u>WildPackets</u> EtherPeek VX monitor <u>OOS</u>, packet loss, voice quality, etc.
- See Also:
 - Call Quality Metrics
 - Network World Review of 7 VOIP Analysis Tools

Alumno: