NORTEL NETWORKS

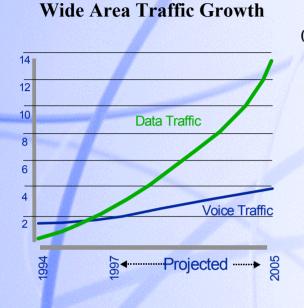
Driving VoIP Solutions

The Reality of the Technology and the Market

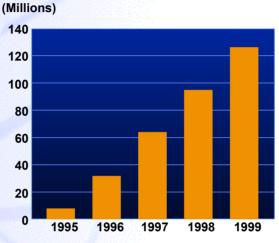
Phil Edholm, CTO and VP, Network Architecture Enterprise Product Portfolio

NORTEL NETWO

Drivers of Change



Worldwide Internet Users



Source: International Data Corporation

Switched L2/L3

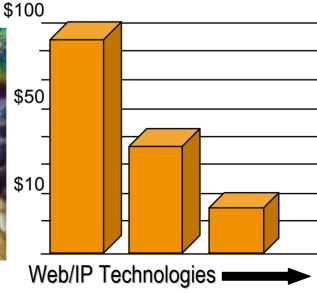


Business Costs





Deregulation



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What's IP got to do with it?

- It's open and ubiquitous
- It's connectionless
- It's not sensitive to distance, time or usage
- It rides Moore's & Metcalf's laws
- It's transport independent
 - It's application independent
 - It's potential is unlimited
 - It's getting better at QoS



Why IP Telephony is good for customers?

Hard Dollar cost reduction

- WAN Cost Reduction
- Improved WAN Price/performance
- Consolidation of multiple networks in one

Operational Improvement

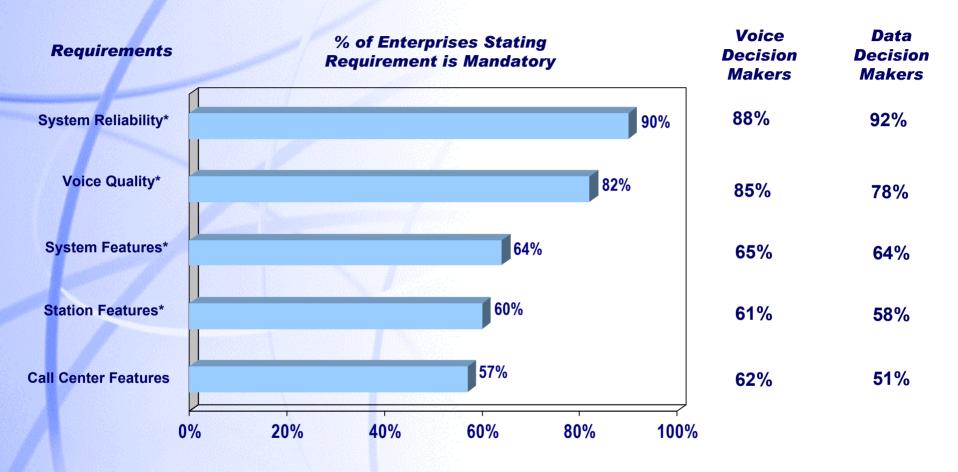
- LAN/Campus Integration
- Simplified moves, adds and changes via IP/DHCP
- Policy and Directory Consolidation

Enhanced Applications

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- Unified Messaging = improved productivity
- Web enabled Call Centers = improved customer care
- Collaborative/ Mobility/Open Applications
- The WWWs 350M users by 2003

Customer Internet Telephony Requirements



* Equivalent to PBX

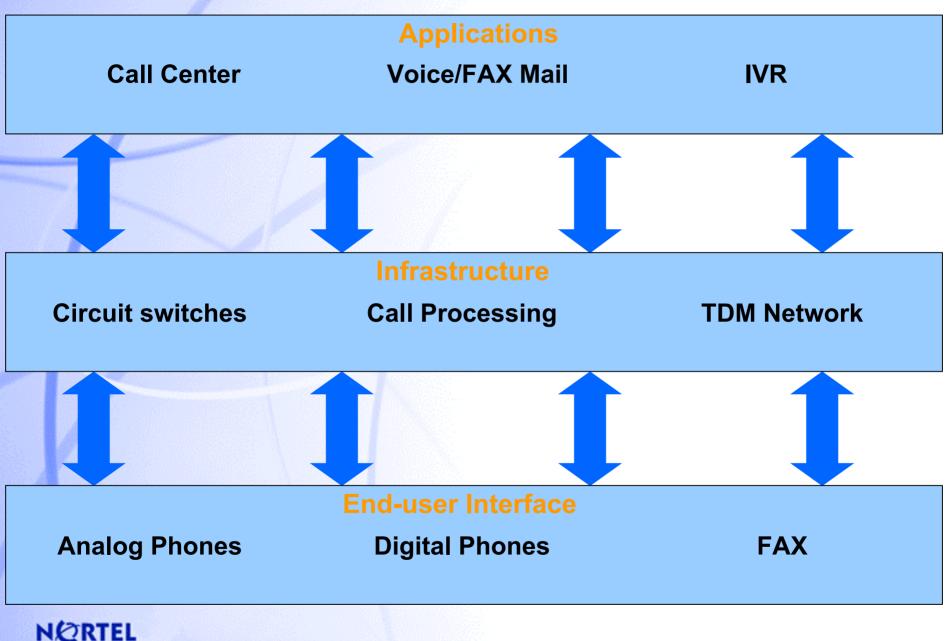
Source: Phillips InfoTech



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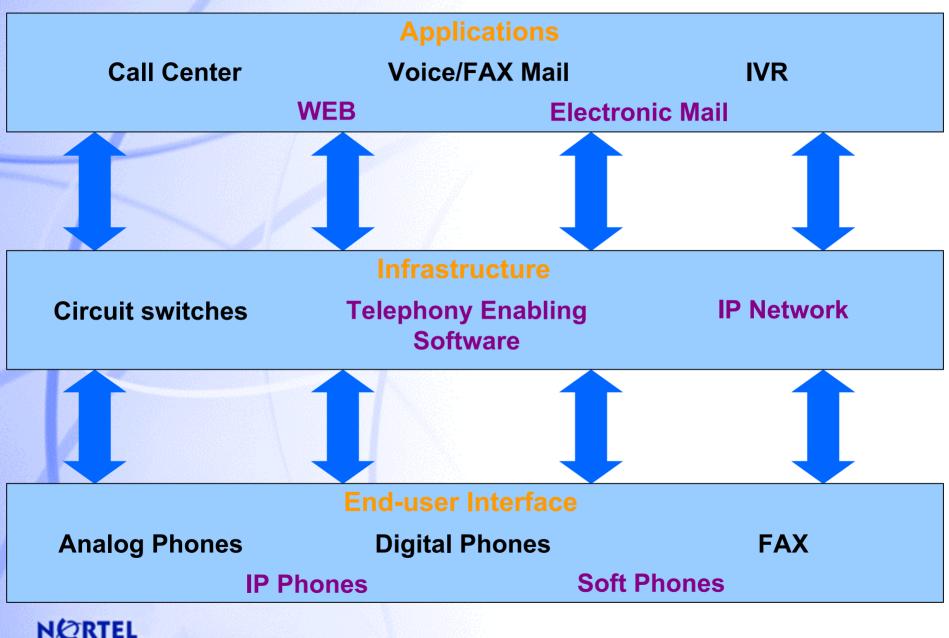
Telephony

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Internet Telephony

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What to look for in Internet Telephony?

- Quality Telephony
- Extensive feature sets & applications
- Global capability
- Bullet-proof reliability
- Low Total Cost of Ownership
- Scalability
- Skills & support availability

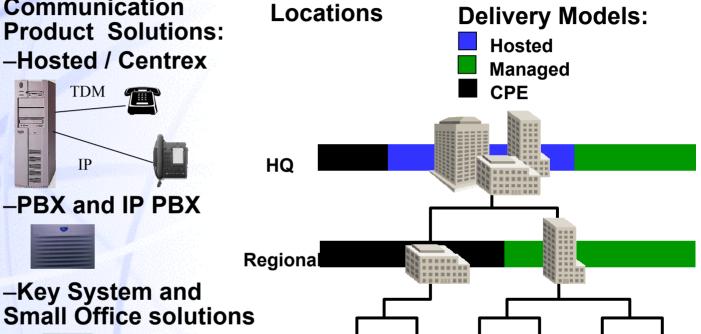


Enterprise Communications

- 1. Various Roles and Services:
- Executives •
- Secretarial •
- Customer • Engagements
- Consulting •
- Sales •
- IT •
- Customer • Service
- **Customer Care** ٠

2. Various Voice 3. Various Communication **Product Solutions:** -Hosted / Centrex TDM THE

IP



0000

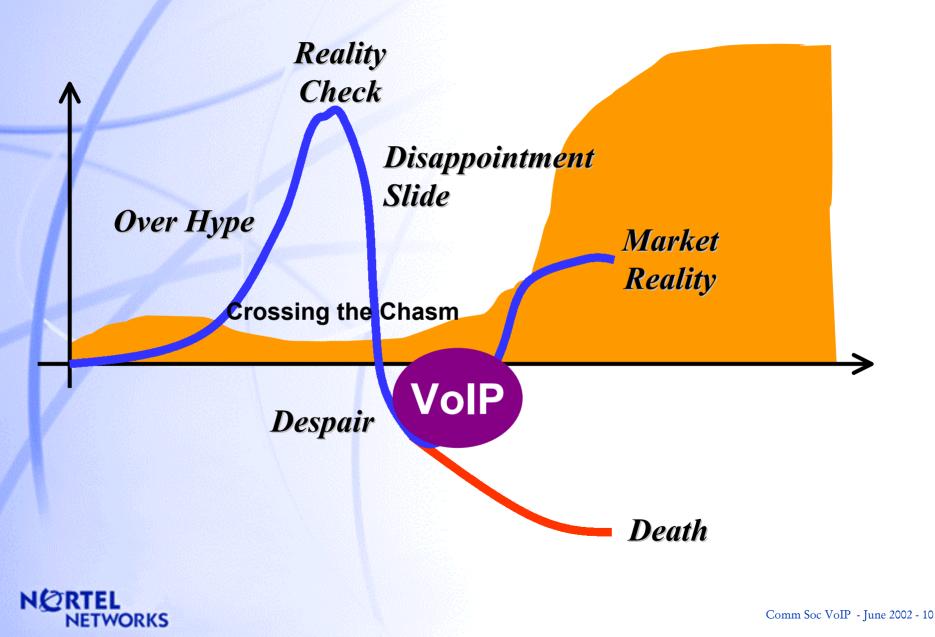
4. Various Business

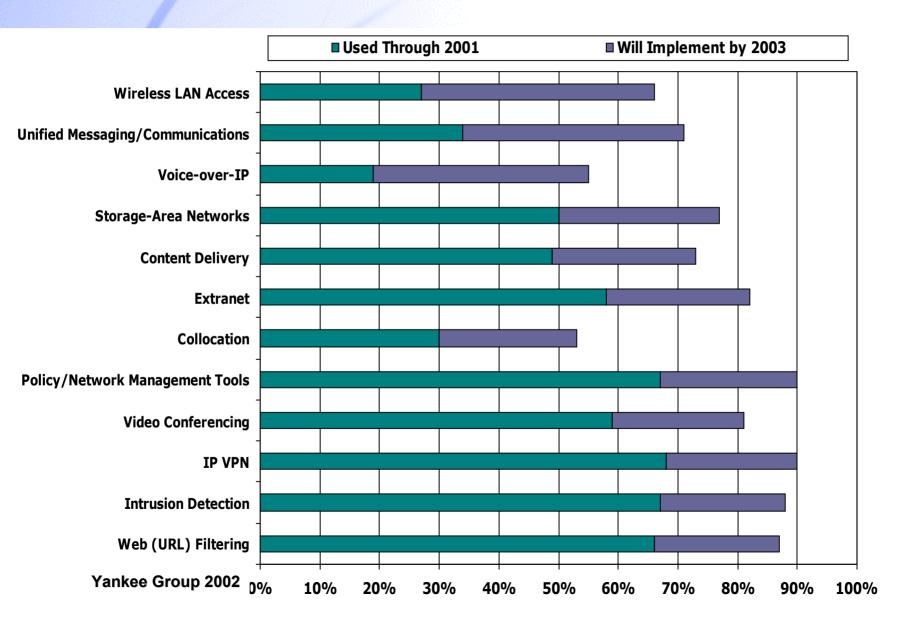
-Dial Access, VoDSL, **Mobility** Remote / Small

Branch

NORTEL NETWORKS **Enterprise Communications Model is not Homogeneous** VoIP migration must address this environment

New Technology Introduction Curve...

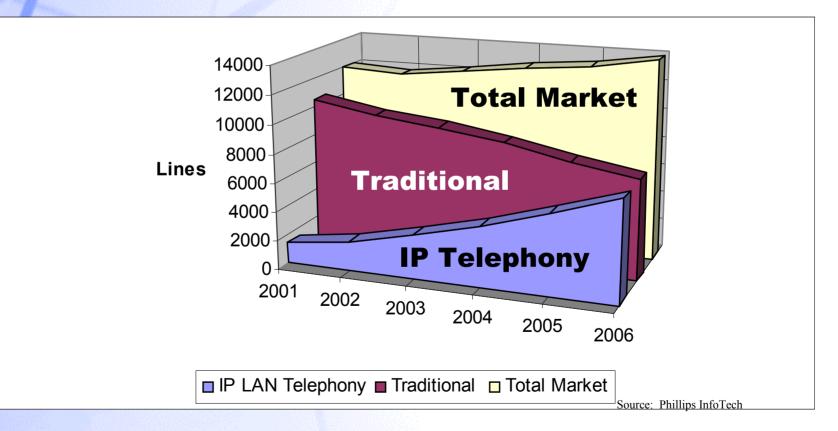




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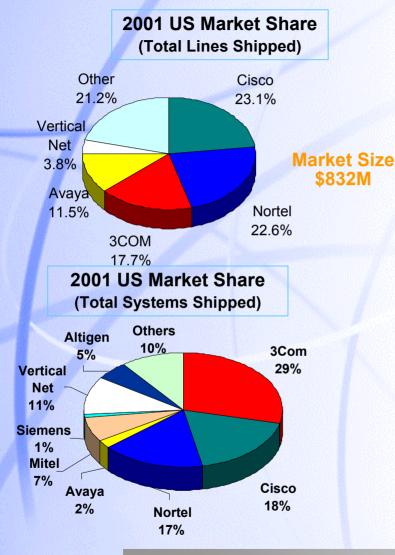
Migration Rate to IP PBXs

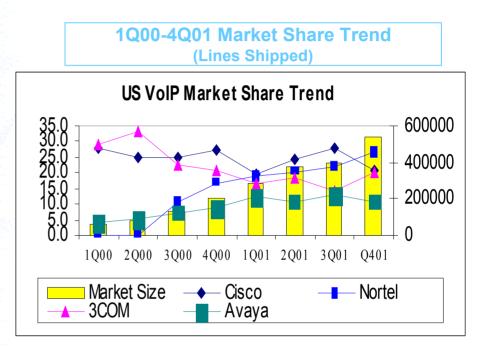
- IP Telephony CAGR at 36%, crossover point for IP and traditional lines in 2006
- The Migration will be Business Case driven not hype drive





2001 US VolP Market Share

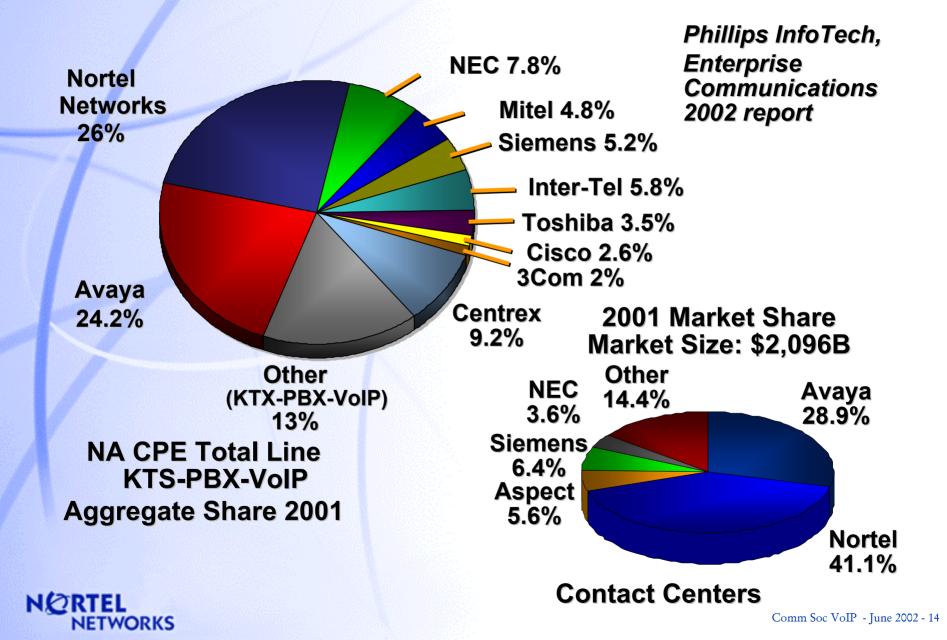




- The Enterprise VoIP market grew another 34.4% Q4'01 over Q3'01, and over 229% year over year.
- Nortel .5% behind Cisco for 2001 total lines shipped (all categories combined)
- Nortel #1 in IP-Enabled segment of VoIP for 2001
- Nortel #1 in Converged segment of VoIP for 2001

Nortel #2 in US Enterprise VolP 2001

North American Voice Market Share



Interoperability and Standards

Control Standards

- H323
 - Megaco
 - SIP

Codec

- Seems to be converging on:
 G729
 - G711 PCM



Summary of Market Data

- Internet Telephony is a reality it will happen
- Knowing how and when is the challenge
- Customers will NOT adopt incomplete solutions
- Installed based TDM transition is a key requirement
- Inter-working is critical
- Common components should be offered
- Standards are needed for openness

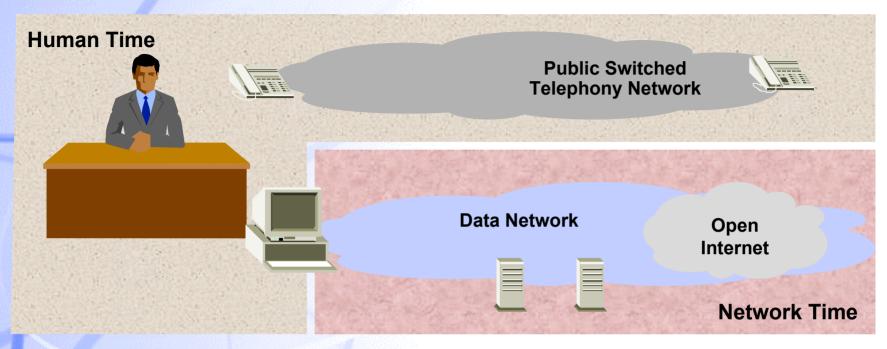


Delivering Voice Services over IP Networks



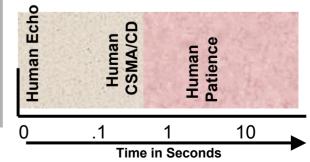
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Human Time Versus Network Time



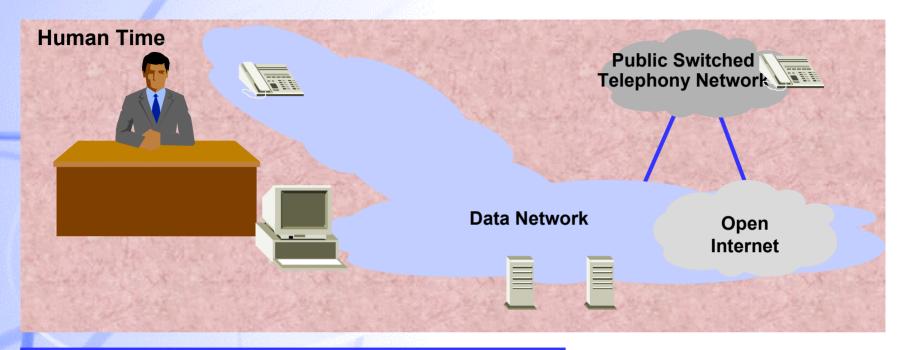
- Human time is based on interpersonal communications and human tolerance for delay
- Human time dominates voice and video communication
- The PC operates in human time to the user and isolates him from network time
- Data network time is for batch functions (file transfer, printing, Web pages, etc.) and is based on patience
- Human interaction "CSMA/CD" defines the edge of human time at about 250-300 msecs





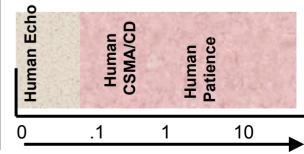


Data networks must operate in human time



- Data networks must support human time in the future
- VoIP and VideoIP require human time
- Collaboration and community applications on the data network have strong interaction component and must operate in human time to succeed
- Network computing and thin clients require human time services
- Policy and QoS are the only way to meld human time onto the data network and maintain the overall capabilities of the system

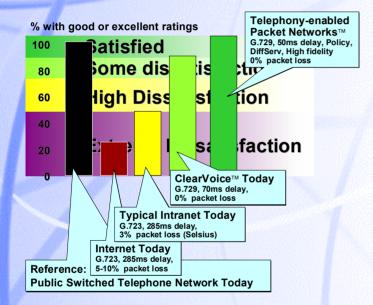
Human Tolerance of Delay





Telephony Session Quality

Nortel Networks test of consumer expectations for voice quality, 1998



Delay Impact on Perceived Voice Quality	
Round Trip Delay	%Poor or Worse
150 msec	3
175 msec	4
200 msec	5.5
225 msec	6.5
250 msec	8
300 msec	10.5



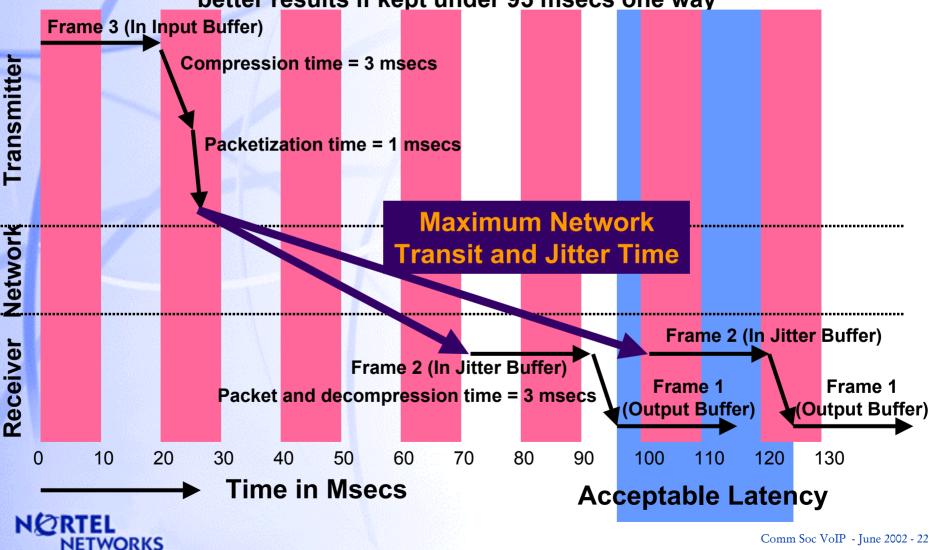
Latency is a sum of the parts

- Quantitization time (10-30 msecs) the sample time to generate a discretely sampled element
- Compression time (2-10 msecs) the time it takes to compress the voice into packet form
- Packetization time (1msec) processing time to assemble and transmit the packet
- Network transit time (variable) the time to transition the network
- Reverse packet and decompression (2-3 msecs) the time it takes on the receiving end to output the samples
- Latency is for a round trip



Time model of network

Round trip time must be less than 250 msecs or 125 msecs one way (voice input to output) to meet criteria for telephony quality interactive voice - better results if kept under 95 msecs one way



Network transit time

- Additional Jitter Packets = 20 msecs each
- Packet Transmission time =
- For 5000 miles = 38 msecs
- Switch or Router hops
 - Switch hops with cut-through < 1 msec</p>
 - Routers can add time 1-3 msecs

Total must be less than 75 msecs at all times

 For a ten router hop net with two extra jitter packets and a 5000 mile trip, the total is: 98 msecs

Distance

8*Sol

Packet Length

1x speed

Keys to successful voice is minimizing jitter packets through QoS/CoS, and grooming the network for router hops



Lost Packets

- Each packet represents about 20 msecs of voice
 - A single lost packet is noticeable (there is no time for error correction)
 Multiple packets lost (or underruns) is interpreted as bad quality
 Packets are lost due to errors or congestion
- Typical Bit Error Rates of current networks generate acceptable performance
 - BER = 10^{-8} (100,000,000 bits transmitted without an error)

Assume total of transited hops is 10

 10^8 bits

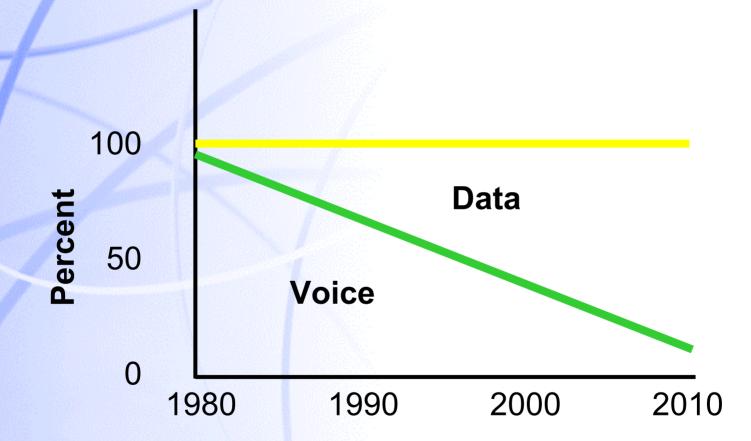
= 11 minutes between occurrences

(10 hops)(15000 bps * 60s/m)

Reliable BER generates acceptable voice, lost packets due to congestion (no QoS queuing) will dramatically degrade quality as will high BER networks paths or circuits

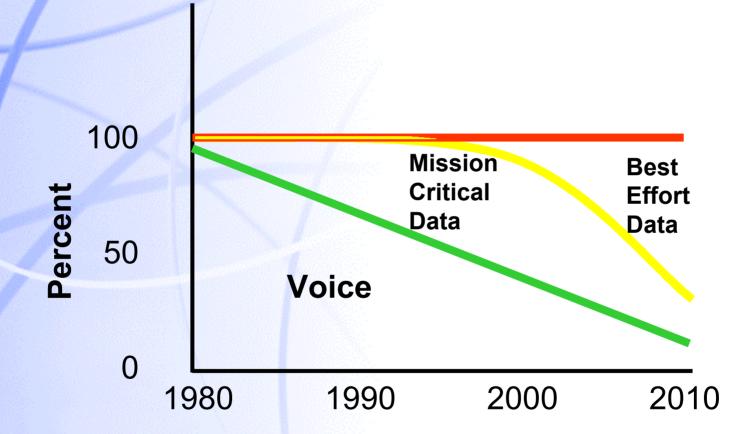
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Driving Class Based QoS



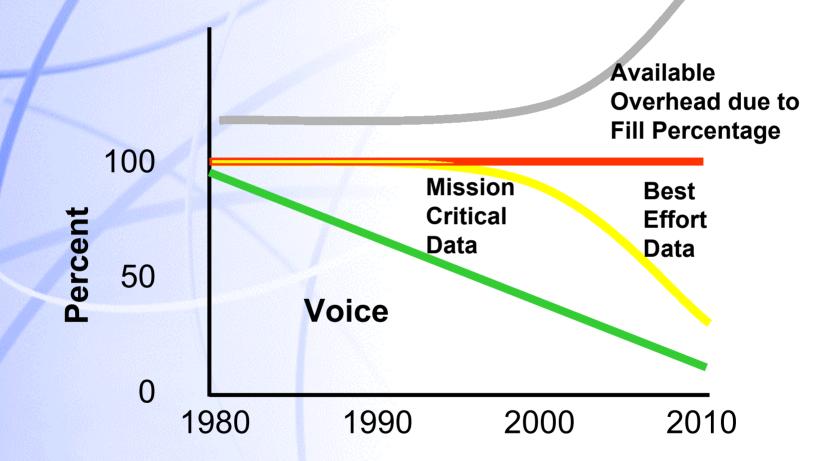


Driving Class Based QoS

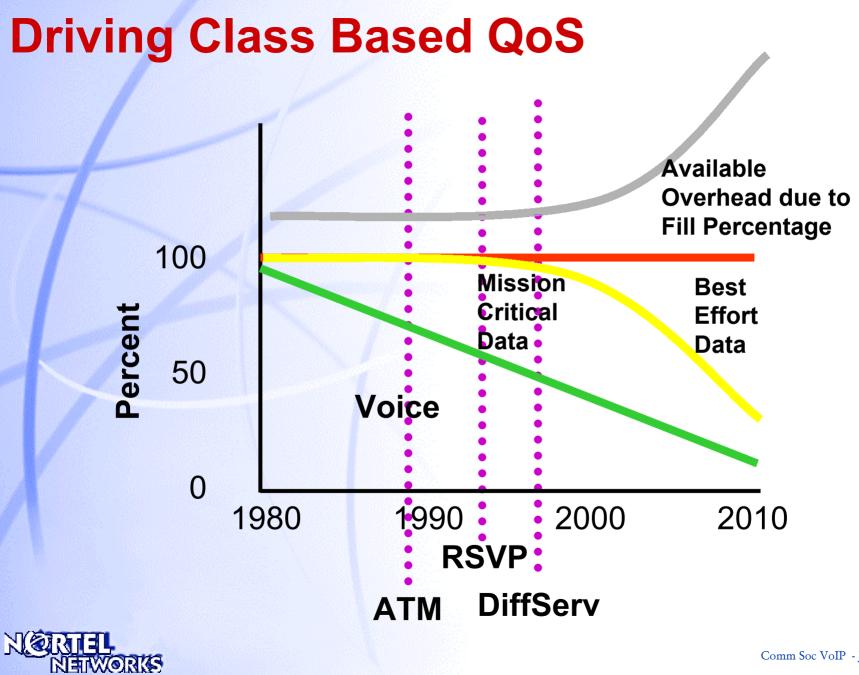




Driving Class Based QoS

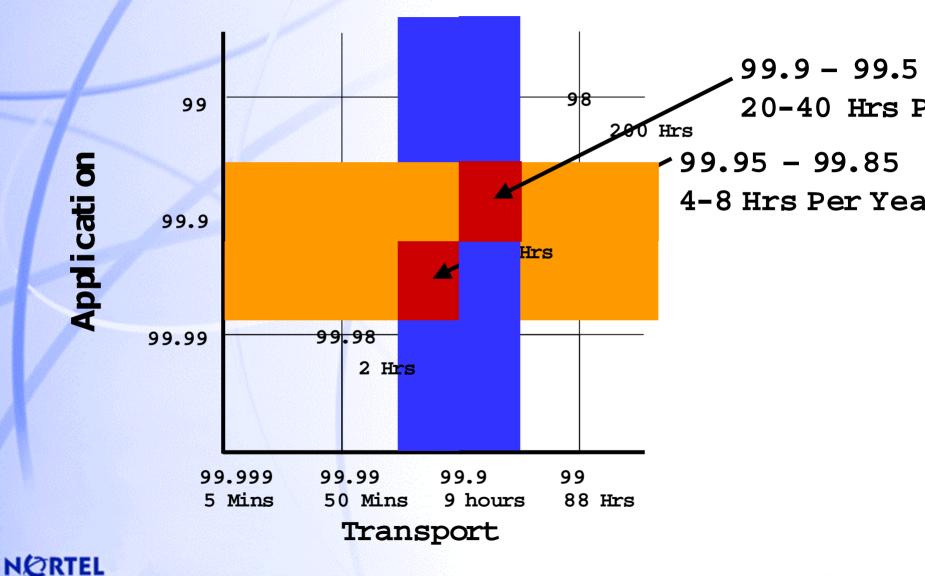






Reliability Transformation

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Decision Values of VolP

- Control Integration
- Media Integration
- Cost



VoIP Control Integration

- Will be the ultimate driver
- Application Integration
- Currently based on TAPI

 Just like a PBX
- Future is web paradigm



SIP Integration Multimedia Collaboration

- Multimedia Services
 - Videoconferencing
 - Instant Messaging
 - Audio & video streaming

Telephony Services

- Call redirect
- Multiple server registration
- Real time call management

Mobility

- Dynamic registration, Findme/Follow-me
- Hoteling

Personalization

- Personal Call Agent
- Dynamic Call Handling

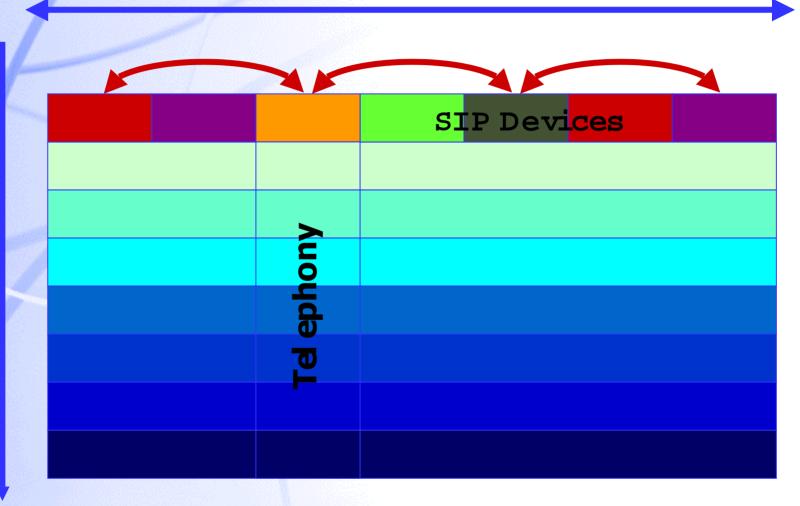


SIP and Telephony



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VoIP Media Integration

- Single wire value
- Not typical VoIP is emerging as a separate device from the PC
- Media integration values
 - Recording

Better as Server Functions

Transforming

Synthesizing

- As voice is person to person, media alterations reduce value
 - Potential for handicapped access and dating
- Future values of packetized voice will emerge



New Media Capabilities

New Conferencing Paradigm

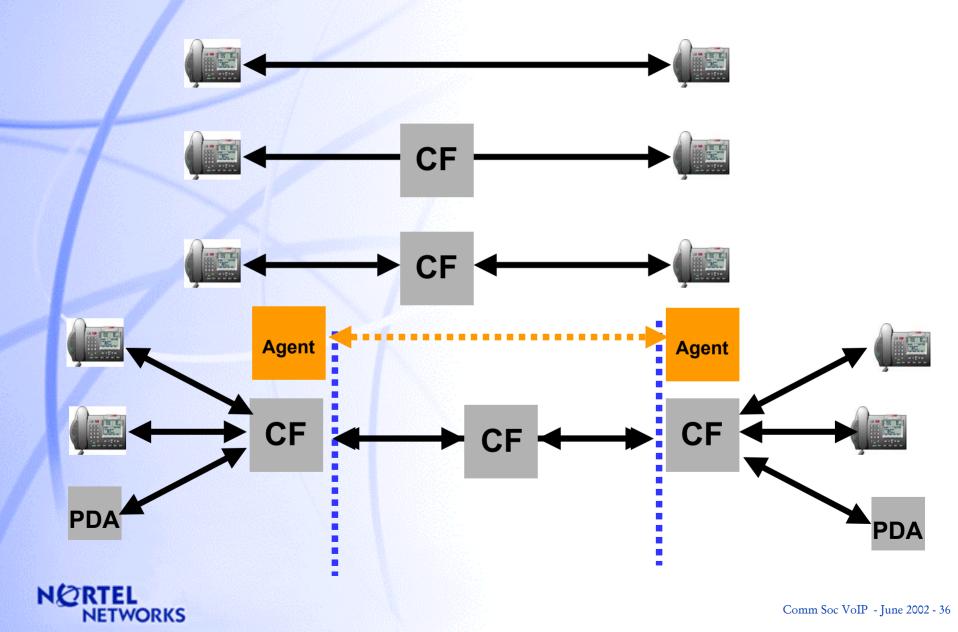
- -500 microsec added latency
- 10-100K conference ports on a PC
- —Fungible in growth

New Capabilities

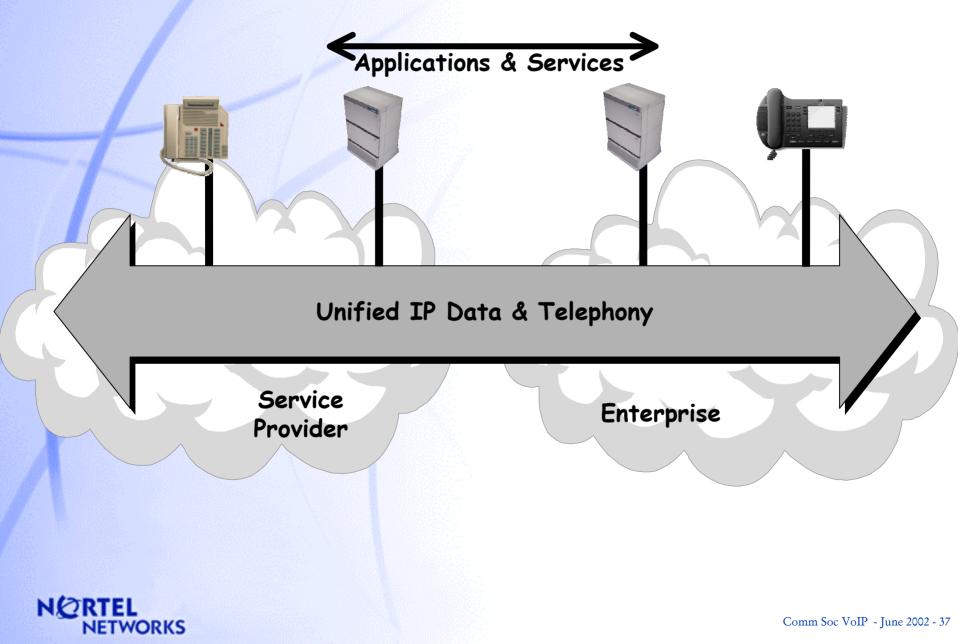
- —Constant Conferencing
- —Every Call includes a conference call
- —Audio "Virtual Space"
- —Integrated Personal Agent technology



Redefining Inter-human Communications



Think of telephony like we think of the Web



What is it like?

- "Phone calls can be URLs"
- A conference call is an object... you can "E-mail me a conference call"
- Telephony apps are buttons on Web pages
- Voice control panel is Web browser
- Like the Web, telephony value & content can be anywhere
- Streaming Web audio/video is now part of telephony
- Service creation is rapid
- This vision drives IP Enabled and Succession



VoIP Cost Impact

Acquisition cost

VoIP equipment
 IP capable network

Ownership cost

- MAC (Moves, Adds, and Changes)
- Support
- Other functions
 - 911 tracking

Note: all costs are in US Dollars



Desktop Cost Comparison





\$1 per day Nortel/Customer Analysis



Typical IT Dept spends less than 3% of budget on voice

Now VoIP?

- Intelligent Edge
- Distributed Intelligence
- Increased Support

Computing Trends

- Network Computing
- Network Displays
- Centralized models

Gut Feel - is this really going to be less?



Long Distance - the real value?

- Assume 1 hour per day per employee
- Assume 50% is intra-business (within the company)
- Current long distance at under \$.05 going to 2 cents
- Cost of long distance at 2 cents is \$288 per employee per year, \$144 for intra-business
- Assume one pair of VoIP ports for every 3-5 employees

 Trunk and line or trunk
- At \$600 per port, and all intra-business long distance on IP and no charge for bandwidth:
- Payback is about 2-3 years

International Traffic is the Real Value



Is it really free?

Voice will be free in a few years

A few decades hence, energy may be free - just like unmetered air. John von Neumann, <u>1956</u>



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What if it is really free?

- Assume that the VoIP equipment is free
 - All telephones, gateways, gatekeepers, call processors, etc.
 All maintenance and support
- But You have to live with.....

Data reliability of 99% versus Voice Reliability of 99.999%

In other words, telephones go from 5 minutes to 88 hours per year of unavailability to your customers



The real cost of reliability loss

- Assume that of the 88 hours, 30 occur during business time
 - Average business is open 6x9 = 2815 hours per year
 - Total hours = 8760
 - Average open is 32%
 - 32% of 88 is 28 hours round up because of Murphy's Law
- Assume \$300,000 revenue per employee with 50% gross margin
- Lost revenue = 30hrs/2815hrs * \$300K = \$3197 per year
- Assume 50% margin margin/earning loss is = \$1598 per year



To Save.....



\$1 per day about \$250 per year

....the business lost over

\$1598 per year!

And the CIO lost his job....



Defining the Future

Greater Than 100 Line Solutions

Nortel **"Nortel** leads **C**isco in both Avaya Ability to Siemens Vision & Execute Mitel • Ο NEC 3Com **Ability to** Alcatel Execute." Shoreline Completeness of Vision

SOURCE: Gartner Research, February 2002.



Nortel Networks Leadership

#1 CPE North America (Total Lines Shipped) - InfoTech #1 VoIP US Q4 '01 (Total Lines Shipped) - InfoTech, The Eastern Management Group #1 VoIP US Q4 '01 (Total Systems Shipped) - InfoTech **#1 VoIP IP-Enabled US (Total Lines Shipped) - InfoTech #1 VoIP Converged US (Total Lines Shipped) - InfoTech** #1 KTS North America (Total Stations Shipped) - InfoTech #1 KTS Canada (Total Stations Shipped) - InfoTech **#1 PBX Canada (Total Lines Shipped) - InfoTech** #1 in IP phone systems Q4 '01 - (The Eastern Management **#1 Multiservice WAN Switch Q4 '01 - SRG, Dell'Oro, In-Stat #1 Enterprise WAN Switch Q4 '01 - SRG, Dell'Oro, In-Stat** #1 ATM Core Switch Q4 '01 - Infonetics **#1 IP Service Switch Q4 '01 - SRG, Infonetics**

Enterprise Solutions Nortel Networks Succession Strategy



Succession Communications Server for Enterprise 1000

Internet Enabled Traditional Voice Portfolio

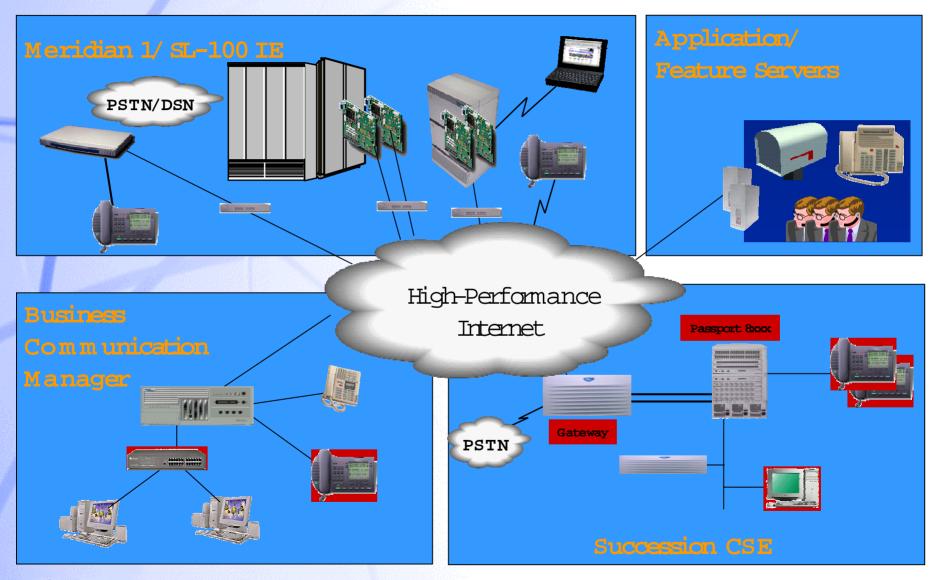
BuildOut Voice Values to Enterprise Network & Managed Services Emerging Market



CSE1K Cal

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Nortel Networks IP Telephony Portfolio



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Succession Enterprise

N



CSEMX brings it all together and completes our strategy

Introducing CSE MX

•CSE <u>Multimedia</u> <u>X</u>change is a new VoIP Telephony Applications Platform

Multimedia applications •Collaborative, mobility,presence, Video services, personal call Management •Xchange •VPN services, number plan consolidation, H. 323, MGCP,

STP

NORTEI

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0

•Built in the Java program ming environment

> •Hardware independent

> > SIP Applications development engine

•CSEMX has evolved from IMS &CS 3000

•Service provider heritage brings ultimate reliability



Conclusions - as if you weren't there already

- Nortel is the leader in Voice and Voice over Data, including IP
- We are firmly committed to IP based solutions
- VoIP needs to be implemented for the right reasons
- Convergence and applications values will drive the implementations
- Be careful of flimsy, poorly thought savings schemes that do not reflect reality
- VoIP will be driven, as the web was, not by IP, but by transformational communications value

