

Broadband Quality of Service

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- The users
- The QoS metrics
- The evaluation methodology
- ITU-T 1540/1541
- Summary

The Next Billion



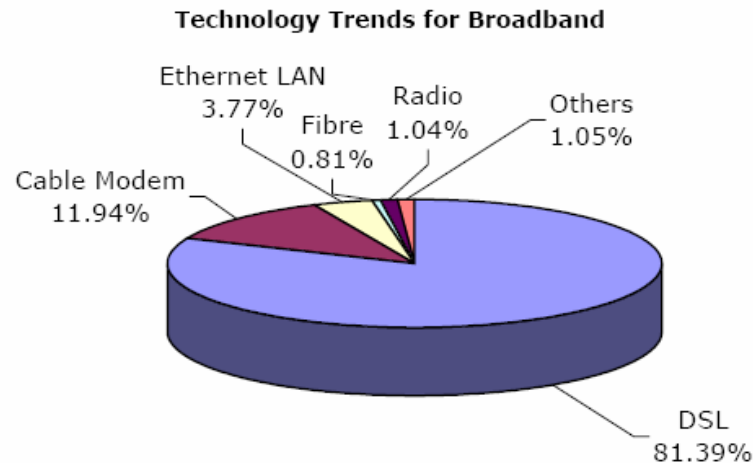
Required Coverage
 ~70% of population
 4900 Towns ~300M people
 ~350 000 Villages ~450M people



Existing Coverage
 ~30% of population
 1700 Towns ~200M people
 Negligible rural coverage

Emerging Market Explosion

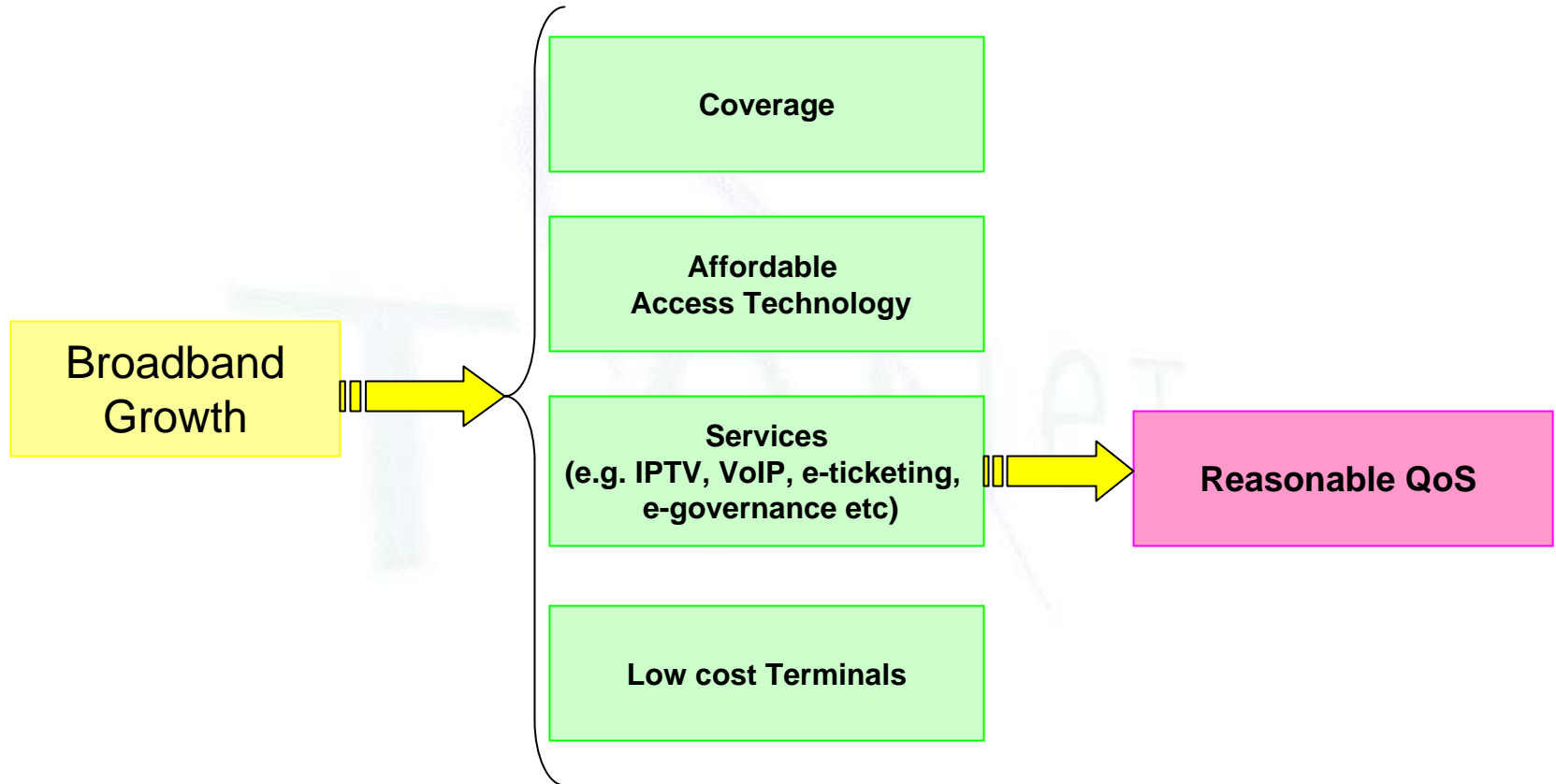
- India adds **10M** telephone lines monthly: crossed 250M in 2007
 - Was struggling 12 years ago to add 1M lines yearly
 - Right price point exploded the market
- Government policy to add **50M** Broadband connections in 4 years
 - Today only 2.5M Broadband connections



Similar trend in other Emerging Markets

- Dense urban areas – 10,000 people/sq. km
- Sparse rural areas
 - Village every 3-10 km
 - Population 1,000 - 25,000
 - Within 25 km of fibre
- ARPU today for 2.5M subscribers : Rs. 280 (\$7)
- ARPU tomorrow for rapid expansion: Rs. 100 (\$2.5)

Enablers for Broadband Growth



- ITU-T: 1.5 Mb/s
- TRAI : 256 kb/s
- Subscriber:
Good experience with common services -- multi-media browsing, downloads, streaming media, VOIP, multi-player games
Requires 256 kb/s - 1 Mb/s
- QoS Metrics Standardization
 - ITU-T 1540/1541
 - IETF IPPM

ITU-T 1540	Remarks
Throughput	Defined in ITU-T Y.1221
IPTD	IP packet Transfer Delay
IPDV	IP packet Delay Variation
IPLR	IP packet Loss Ratio
IPER	IP Packet Error Ratio
SIPR	Spurious IP packet Ratio
Availability	<ul style="list-style-type: none">• Y.1540 defines availability if $IPLR < 0.75$.• Minimum observation period of 5 min.• Defined as Percent availability

IPPM	Remarks
Link/Path Bandwidth Capacity RFC5136	Capacity of the Link/Path
Bulk Transport Capacity RFC3138	Bandwidth available at transport layer
One-way & Two-way Delay RFC2679 , RFC2681	Delay experienced by the packet from source to destination (and back)
Delay Variation RFC3393	Variation in Delay
Packet Re-ordering RFC4737	Number of packets received out of sequence in the Destination
One-way & Two-way Loss RFC2680	Number of packets lost from source to destination (and back)
Loss-pattern RFC2680	Number of packets lost from source to destination (and back)
Connectivity RFC2680	Number of packets lost from source to destination (and back)

Params	Class-0	Class-1	Class-2	Class-3	Class-4	Class-5
IPTD	100 ms	400 ms	100 ms	400 ms	1 s	U
IPDV	50 ms	50 ms	U	U	U	U
IPLR	1×10^{-3}	1×10^{-3}	1×10^{-3}	1×10^{-3}	1×10^{-3}	U
IPER	1×10^{-4}	1×10^{-4}	1×10^{-4}	1×10^{-4}	1×10^{-4}	U

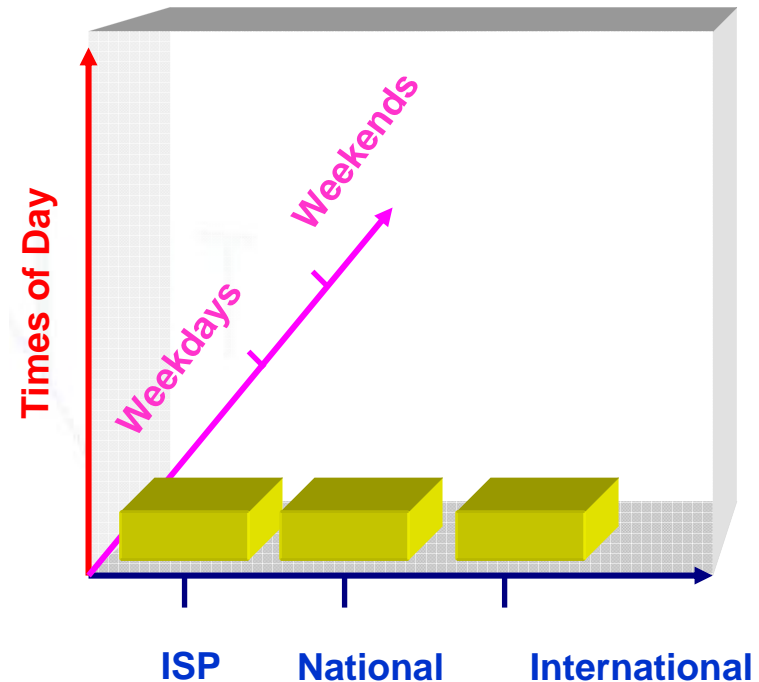
- More Subscriber Oriented Metrics, objective way of measuring subscriber QoE
 - Download throughput
 - Download Bandwidth available to the user
 - Upload throughput
 - Upload Bandwidth available to the user
 - Round-trip delay (RTT)
 - Time taken for a packet to reach a destination and return
 - Delay jitter
 - Average variation in RTT
 - Packet loss
 - No of packets lost, expressed in %
 - Availability of service

Comparison with ITU-T & IETF

Our Model	IETF IPPM	ITU-T
Throughput (upload & Download)	<ul style="list-style-type: none"> • Link/Path Bandwidth Capacity • Bulk Transport Capacity 	ITU-T Y.1221
Delay (RTT)	One-way & Two-way Delay	IP packet Transfer Delay (IPTD)
Jitter	Delay Variation	IP packet Delay Variation (IPDV)
Packet Loss	<ul style="list-style-type: none"> • Loss-pattern • One-way & Two-way Loss 	IP packet Loss Ratio (IPLR)
		IP Packet Error Ratio (IPER)
	Packet Re-ordering	
		Spurious IP packet ratio (SIPR)
Availability	Connectivity	Defined with IPLR (IPLR < 0.75)

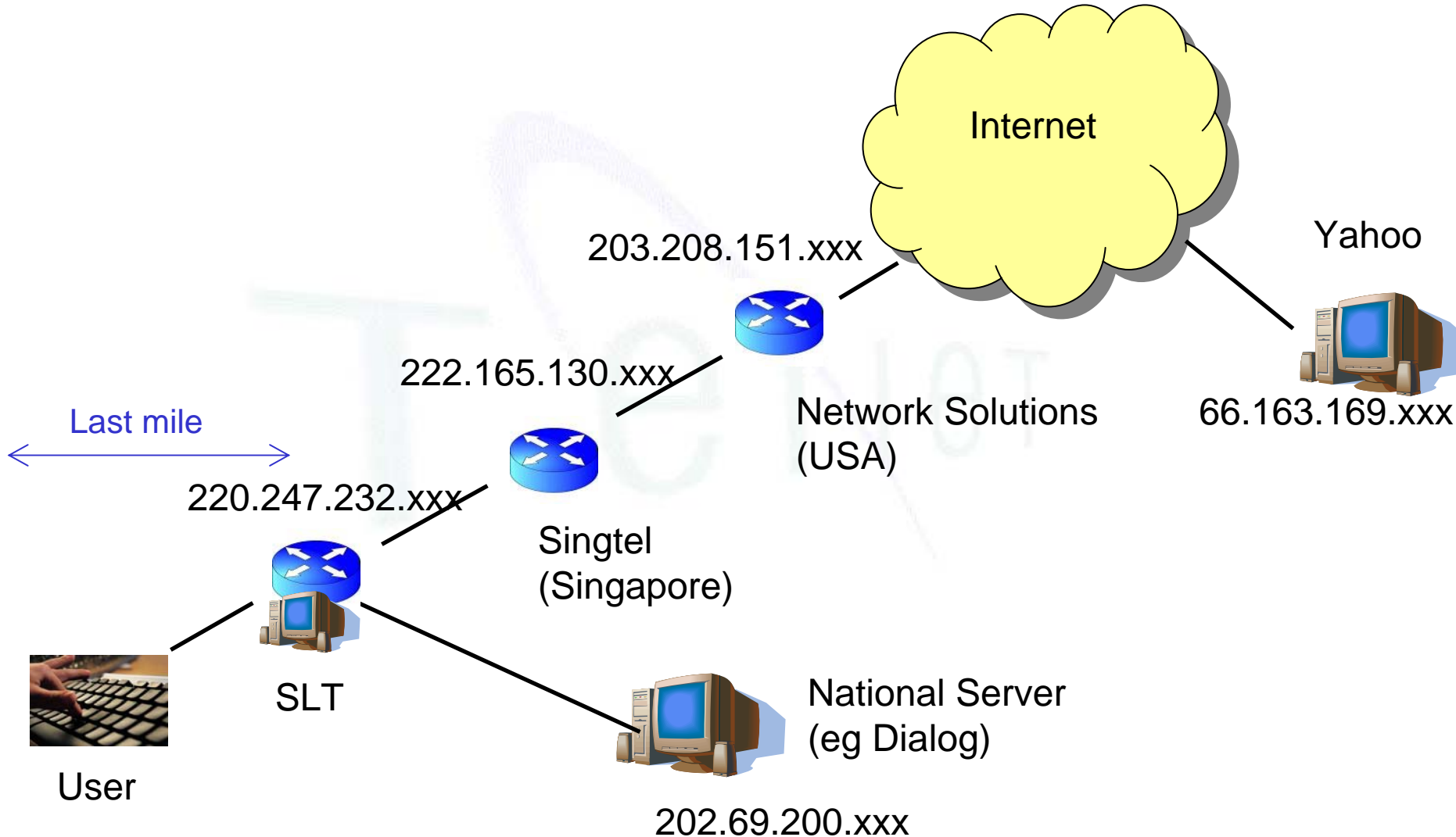
Test Methodology (1)

- Tests 6 parameters
 - Covers most aspects of subscriber QoE
- Tests three servers (ISP, National, International)
 - ISP local N/w, National Peering and International Bw
- Repeated at different times of the day
 - Provisioning & Dimensioning of N/w
- Repeated at weekdays and weekends
 - Provisioning & Dimensioning of N/w

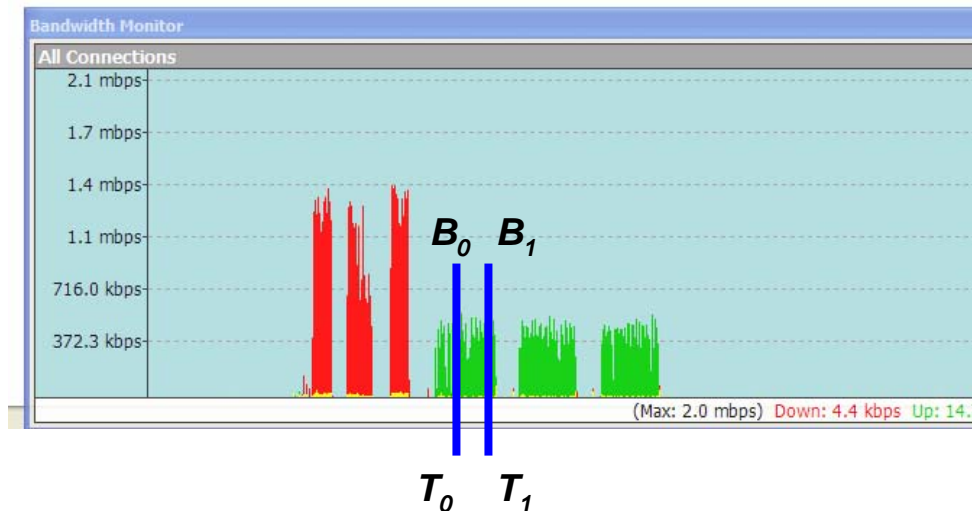


- Uses commonly available tools
 - BW monitor
 - Ping
 - Tracert
- Tests for long intervals to minimize effects of short term variations (e.g. 100 pings, 100 sec download)
- Variations studied and outliers removed

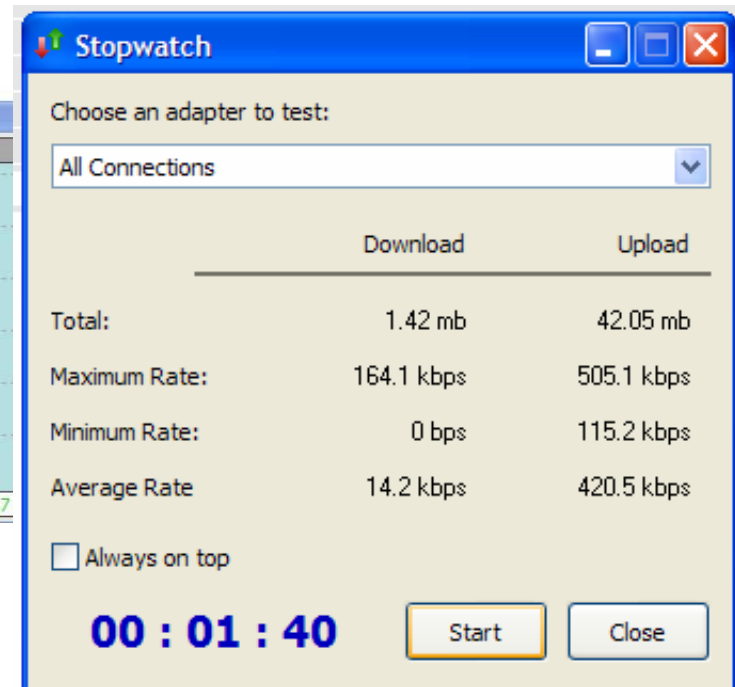
Network Diagram of a Test



- Download/Upload a large file (~ 5 MB)
- Throughput Measurement
 - A. File size / Elapsed time (Displayed by file transfer utilities)
 - B. More Accurate Result

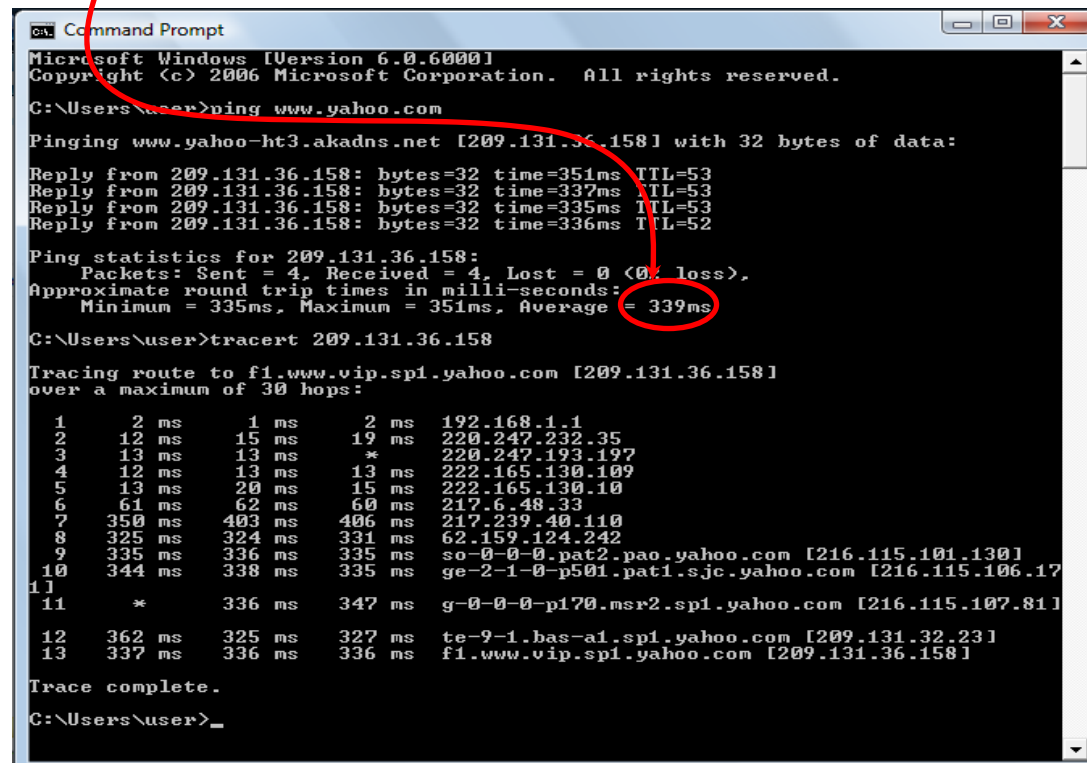


$$\text{Throughput} = (B_1 - B_0) / (T_1 - T_0)$$



Round Trip Time (RTT)

- **RTT**: Time taken for a packet to reach the destination and return.
 - Average RTT (**M**) measured using Ping with 100 packets



```
Command Prompt
Microsoft Windows [Version 6.0.6000]
Copyright (c) 2006 Microsoft Corporation. All rights reserved.

C:\Users\user>ping www.yahoo.com

Pinging www.yahoo-ht3.akadns.net [209.131.36.158] with 32 bytes of data:

Reply from 209.131.36.158: bytes=32 time=351ms TTL=53
Reply from 209.131.36.158: bytes=32 time=337ms TTL=53
Reply from 209.131.36.158: bytes=32 time=335ms TTL=53
Reply from 209.131.36.158: bytes=32 time=336ms TTL=52

Ping statistics for 209.131.36.158:
    Packets: Sent = 4, Received = 4, Lost = 0 (0% loss),
    Approximate round trip times in milli-seconds:
        Minimum = 335ms, Maximum = 351ms, Average = 339ms

C:\Users\user>tracert 209.131.36.158

Tracing route to f1.www.vip.sp1.yahoo.com [209.131.36.158]
over a maximum of 30 hops:
  0  1 ms    1 ms    1 ms    192.168.1.1
  1  12 ms   15 ms   19 ms   220.247.232.35
  2  13 ms   13 ms   *      220.247.193.197
  3  12 ms   13 ms   13 ms   222.165.130.109
  4  13 ms   20 ms   15 ms   222.165.130.10
  5  61 ms   62 ms   60 ms   217.6.48.33
  6  350 ms  403 ms  406 ms  217.239.40.110
  7  325 ms  324 ms  331 ms  62.159.124.242
  8  335 ms  336 ms  335 ms  so-0-0-0.pat2.pao.yahoo.com [216.115.101.130]
  9  344 ms  338 ms  335 ms  ge-2-1-0-p501.pat1.sjc.yahoo.com [216.115.106.17]
 10 *      336 ms  347 ms  g-0-0-0-p170.msr2.sp1.yahoo.com [216.115.107.81]
 11 362 ms  325 ms  327 ms  te-9-1.bas-a1.sp1.yahoo.com [209.131.32.23]
 12 337 ms  336 ms  336 ms  f1.www.vip.sp1.yahoo.com [209.131.36.158]

Trace complete.

C:\Users\user>
```

Jitter and Round Trip Time

- **Jitter**: Variation in RTT

$$k = n$$

$$\sum |M - r_k| / 100$$

$$k = 1$$

M = Average RTT; n = sample size = 100;

r_k = k^{th} RTT reading

- Measured by pinging 100 packets to destination

Packet Loss and Availability

- **Packet Loss** = Number of packets (in %) which do not reach the destination
 - Measured by Ping
- **Availability** = $1 - \text{Prob}[\text{Service unavailable for } >30 \text{ sec}]$
 - Service Unavailable > 30 sec, measured by continuous failure of ping packet for greater than 30 sec.

Service	Throughput		Delay		
	Down	Up	RTT	Jitter	Loss
Browse (text)	++	-	++	-	-
Browse (media)	+++	-	++	+	+
Download file	+++	-	-	-	-
Transactions	-	-	++	+	-
Streaming media	+++	-	++	++	++
VOIP	+	+	+++	+++	+++
Games	+	+	+++	++	++

+++ highly relevant, ++ very relevant, + relevant, - not relevant

- 08:00 hrs: Lean period
- 11:30 hrs: Peak Business hours
- 15:00 hrs: Peak business hours
- 18:00 hrs: Shift from business to residential usage
- 20:30 hrs: Early residential users
- 22:30 hrs: Late residential users

- Tremendous potential for broadband growth
 - Cost of service and QoS are key
- Six objective measures for QoS
- Test methodology to minimize bias and error
- Tests carried out by LirneAsia in Sri Lanka and India

- Design a non-intrusive technique for continuous monitoring of QoS parameters and mechanism for automatic publication
 - Subscriber centric
 - Operator centric
- Class of Service aware QoS Measurement
 - Measure QoS performance for different classes of traffic
 - Bit error tolerant audio codecs for VoIP services

- TeNeT Group – <http://www.tenet.res.in>
- Articles on the various broadband technologies -- <http://en.wikipedia.org>
- Broadband QoS study -- <http://www.lirneasia.net/projects/current-projects/2241/>
- Email: tag@tenet.res.in, rtm@midascomm.com

Thank You