

---

# **Performance & Monitoring Network**

**Muhammad Zen Samsono Hadi, ST. MSc.**

# Functions of Network Management

---

- **F**ault management
  - Network state monitoring
  - Failure logging, reporting and tracking etc.
- **C**onfiguration management
  - device and software configuration
  - version control (compare, apply and rollback, backup) etc.
- **A**ccounting management
  - billing and traffic measurement etc.
- **P**erformance management
- **S**ecurity Management
  - Access control, worm/attack detection and alert etc.

# Performance Management-Why

---

- **Why needed and important?**
  - **Capacity planning**
    - when do we need to upgrade our link and device?
  - **Ensure network availability**
  - **Verify network performance, verify QoS (we expected)**
  - **Ensure SLA compliance (customer expected)**
  - **Better understanding and control of network**
  - **Optimization, make the network runs better!**
- **Proactive or reactive?**
  - **Know problem before users and boss**
  - **Solve the problem before their complain**

Or

  - **Wait for problem to happen, and customers complain?**  
  - **As a NOC, we should be proactive, **NOC** means **NO C**omplain!**

# Performance Management-What

---

- **What's performance management?**
  - understanding the behavior of a network and its elements in response to traffic demands
  - Measuring and reporting of network performance to ensure that performance is maintained at a acceptable level

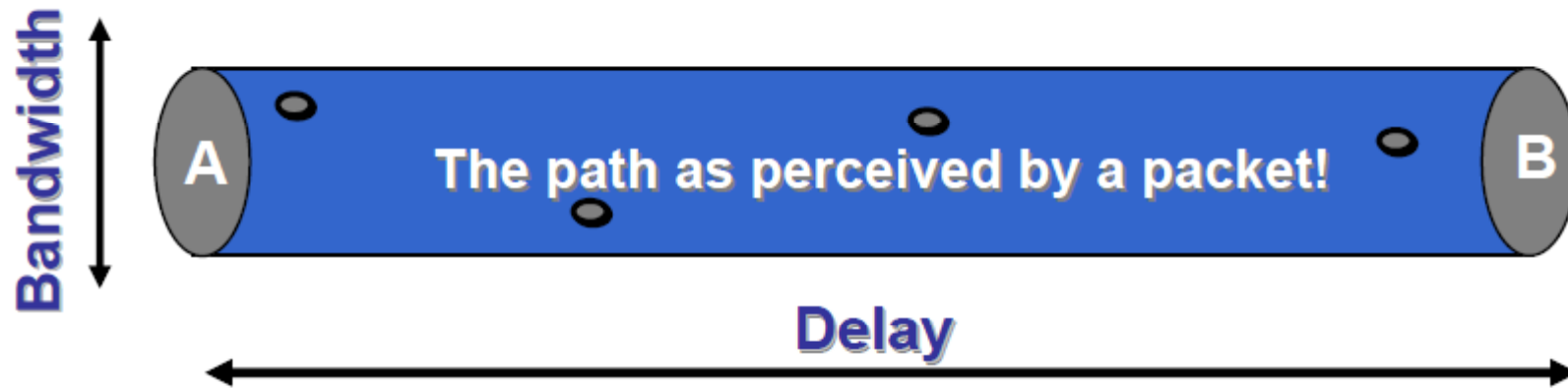
# Performance Management-How

---

- **How to measure the network performance**
  - Delay, jitter, packet loss, bandwidth usage etc.
- **The steps and process of performance management:**
  - Data collection
  - Baseline the network
  - Determining the threshold for acceptable performance
  - Tuning
- **Technologies and tools needed**
  - Data collection technologies such as: sniffing & netflow
  - QoS
  - Tools: ping, mrtg, iperf, wget, etc.

# Analogi Jaringan

---



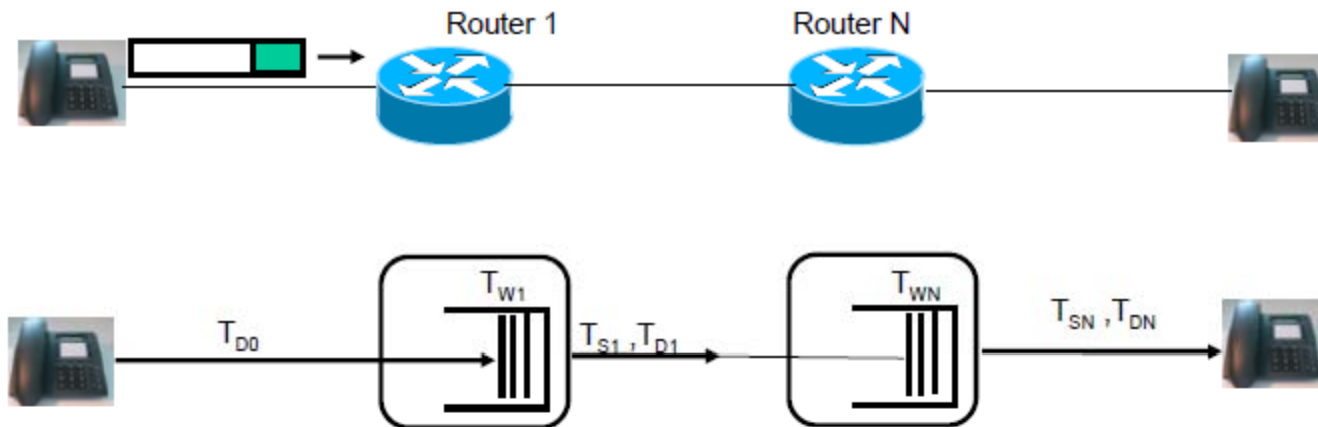
- **Bandwidth, dianalogikan pipe**
- **Delay, mempresentasikan panjang pipa**
- **Jitter, variasi delay pada pipa**
- **Loss, menggambarkan kebocoran pada pipa**

# Delay (Latency)

---

- ***Delay*** adalah waktu yang dibutuhkan oleh sebuah paket data terhitung dari saat pengiriman oleh *transmitter* sampai saat diterima oleh *receiver*
- Delay untuk komunikasi suara :
  - a. Propagation delay*** (*delay* yang terjadi akibat transmisi melalui jarak antar pengirim dan penerima)
  - b. Serialization delay*** (*delay* pada saat proses peletakan bit ke dalam *circuit*)
  - c. Processing delay*** (*delay* yang terjadi saat proses *coding*, *compression*, *decompression* dan *decoding*)
  - d. Packetization delay*** (*delay* yang terjadi saat proses paketasasi *digital voice sample*)
  - e. Queuing delay*** (*delay* akibat waktu tunggu paket sampai dilayani)
  - f. Jitter buffer*** (*delay* akibat adanya *buffer* untuk mengatasi *jitter*)
- Tools: ping, traceroute, tcpdump.

# Perhitungan Delay



- Total delay time of a packet (without CPE) for N links and Routers

$$T_G = T_{D0} + \dots + T_{DN} + T_{W1} + \dots + T_{WN} + T_{S1} + \dots + T_{SN}$$

Delay time of physical lines (5 usec/km)

Waiting time In equipment buffers

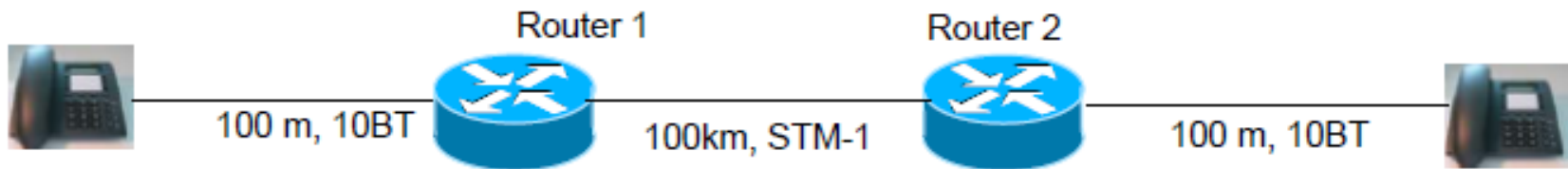
Serialisation time t:  
 $T_s = \text{packet length} / \text{output bitrate}$

- limit for Voice over IP Roundtrip delay  $R = 2 * T_G = 150 \text{ msec}$  (ITU G.113)



# Contoh

---



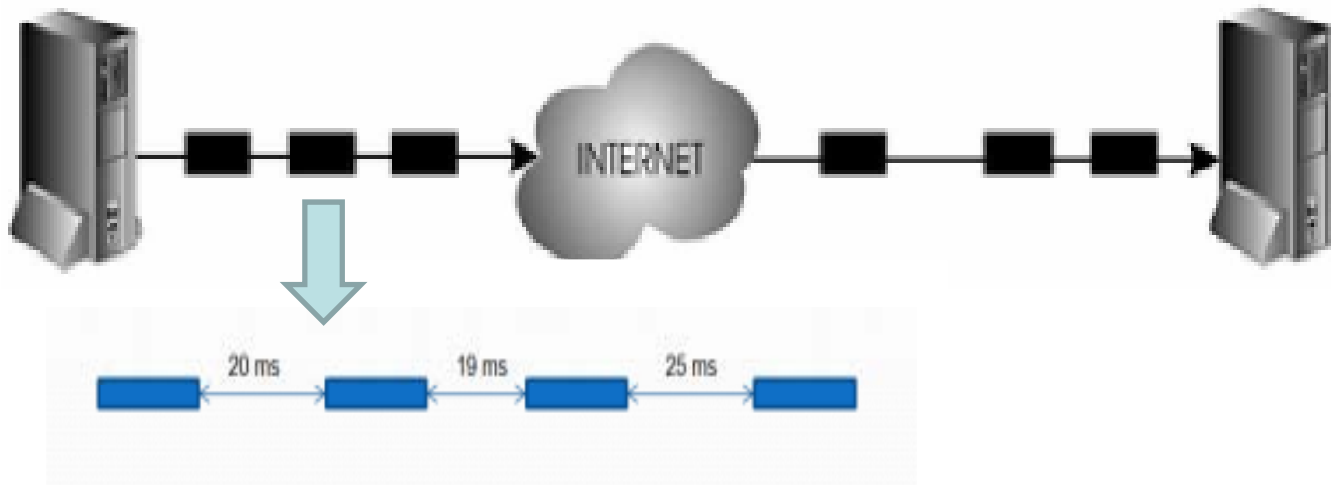
- **VoIP-Phones** dihubungkan dengan 10Mbps ke Router. Panjang paket adalah 212 Byte. Kedua Router dihubungkan melalui WAN dengan STM-1 (155Mbps). Pada masing-masing input dan output Router adalah buffer untuk serialization.

**Waiting time** untuk Router adalah 10msec.

**Berapa total end to end delay (tanpa CPE phone) dan roundtrip delay ?**

# Jitter

- *Jitter* adalah variasi *delay*, yaitu perbedaan selang waktu kedatangan antar paket di terminal tujuan.
- Untuk mengatasi *jitter* maka paket data yang datang dikumpulkan dulu dalam *jitter buffer* selama waktu yang telah ditentukan sampai paket dapat diterima pada sisi penerima dengan urutan yang benar.
- Nilai jitter yang direkomendasikan oleh ITU – T Y.1541 adalah dibawah 50 ms.
- Tools: ping, iperf, dll.  $J1 = \text{abs}(t2-t1)$ ,  $J2 = \text{abs}(t3-t2)$ , ....



# Packet Loss

---

- **Packet loss** adalah banyaknya paket yang hilang selama proses transmisi ke tujuan.
  - Terjadi tabrakan data atau antrian penuh
  - Link atau hardware disebabkan CRC error
  - Perubahan rute (temporary drop) atau blackhole route (persistent drop)
  - Interface or router down
  - Misconfigured access-list
  - ...
- 1% packet loss tidak dapat digunakan.
- **Packet loss** dinyatakan dalam persen (%) dengan nilai yang direkomendasikan pada ITU-T Y.1541 tidak boleh lebih dari 0.1 %.
- Tools: ping etc.

$$\text{Packet loss} = \frac{(\text{Packets}_{\text{transmitted}} - \text{Packets}_{\text{received}})}{\text{Packets}_{\text{transmitted}}} \times 100\%$$

# Throughput

---

- ***Throughput*** adalah jumlah bit yang diterima dengan sukses perdetik melalui sebuah sistem atau media komunikasi (kemampuan sebenarnya suatu jaringan dalam melakukan pengiriman data).
- Throughput diukur setelah transmisi data (host/client) karena suatu sistem akan menambah delay yang disebabkan *processor limitations*, kongesti jaringan, *buffering inefficients*, error transmisi, *traffic loads* atau mungkin desain hardware yang tidak mencukupi.
- Aspek utama throughput yaitu berkisar pada ketersediaan bandwidth yang cukup untuk menjalankan aplikasi.
- Hal ini menentukan besarnya trafik yang dapat diperoleh suatu aplikasi saat melewati jaringan.
- Tool : MRTG, iperf

$$waktu\_download\_terbaik = \frac{ukuran\_file}{bandwidth}$$
$$waktu\_download\_typical = \frac{ukuran\_file}{throughput}$$

# Network Availability

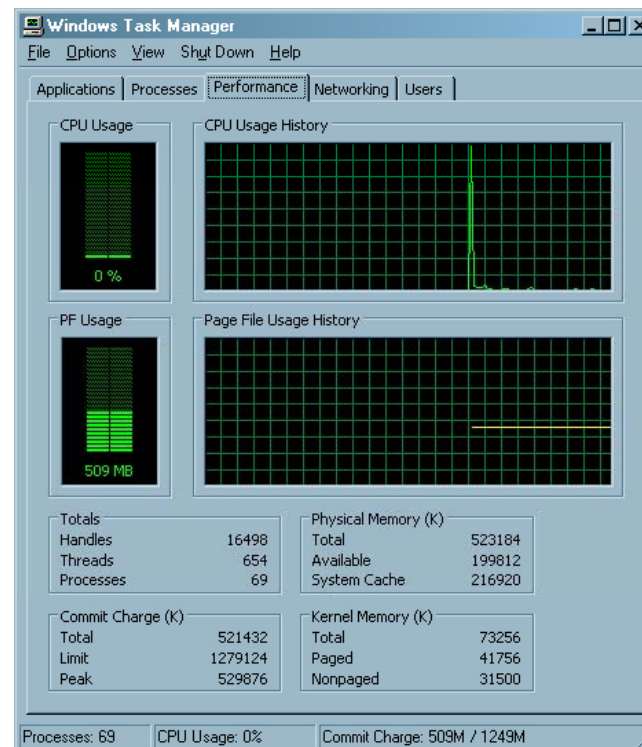
---

- is the metric used to determine uptime and downtime
- $\text{Availability} = (\text{uptime})/(\text{total time}) = 1 - (\text{downtime})/(\text{total time})$
- Network availability is the IP layer reachability
- Better > 99.9%
- 99.9%
  - $30 \times 24 \times 60 \times 0.1\% = 43.3$  (Minutes), means the down time should be less than **45** minutes in one month
- 99.99%
  - $30 \times 24 \times 60 \times 0.01\% = 4.3$  (Minutes), means the down time should be less than **5** minutes in one month!
- **99.9%** is acceptable for R&E networks (Even 99.0% is acceptable), some commercial ISPs can reach 99.99%

# CPU and Memory Utilization

---

- We focus on routers
- CPU utilization better less than 30%
- For global routing routers, at least 512M memory is needed

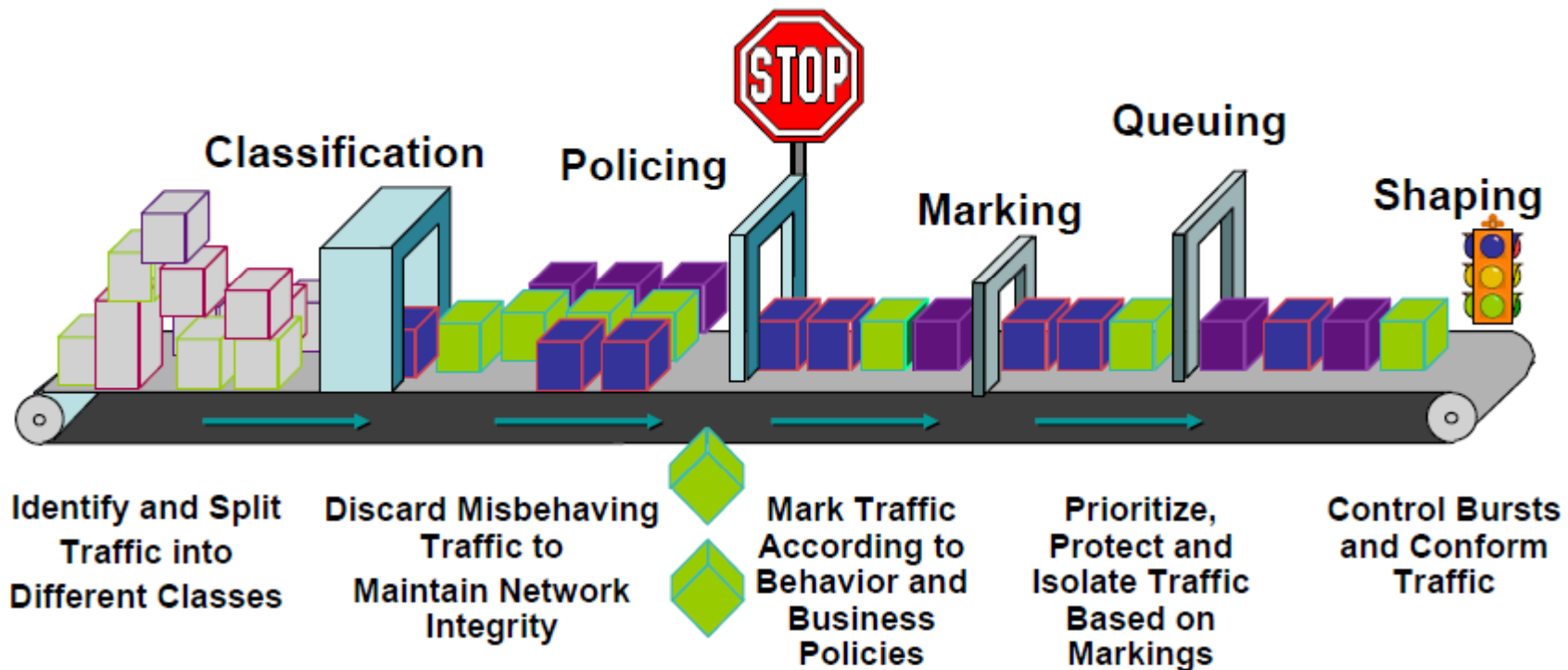


# QoS

---

- **QoS: Quality Of Service**
- **QoS is technology to manage network performance**
- **QoS is a set of performance measurements**
  - **Delay, Jitter, packet loss, availability, bandwidth utilization (throughput) etc.**
- **IP QoS: QoS for IP service**

# QoS Architecture

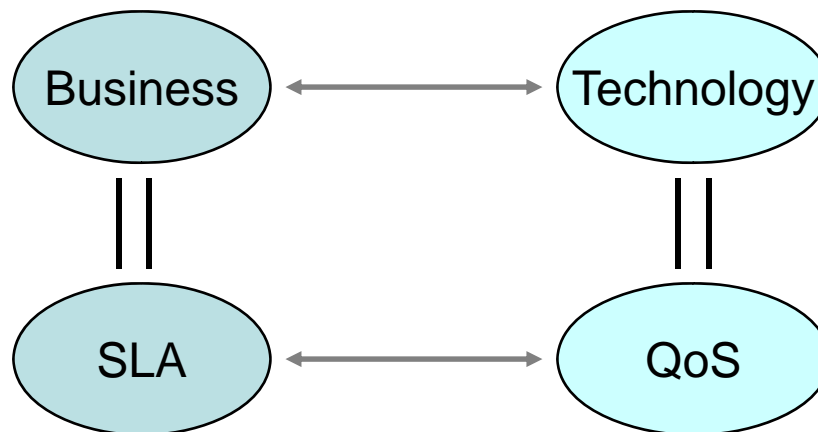




# SLA and QoS

---

- **SLA: Service Level Agreement**
- **SLA is the agreement between service provider and customer, SLA defines the quality of the service the service provider delivered, such as delay, jitter, packet loss etc.**
- **SLA is a very important part of the business contract, and also can be used to distinguish the service level of different ISPs**



# SLA example: Level 3

		<i>Premium</i>	<i>Optimized</i>	<i>Enhanced</i>
<b>Delay</b>	<i>Latency (one way)</i>			
	<i>0 - 250 miles</i>	10ms	10ms	10ms
	<i>251 - 500 miles</i>	15ms	15ms	15ms
	<i>501 - 1000 miles</i>	20ms	20ms	20ms
	<i>1001 - 1500 miles</i>	25ms	25ms	25ms
	<i>1501 - 2000 miles</i>	30ms	30ms	30ms
	<i>2001 - 2500 miles</i>	35ms	35ms	35ms
	<i>2501+ miles</i>	45ms	45ms	45ms
	<i>Trans-Atlantic</i>	45ms	45ms	45ms
<b>Packet Loss</b>	<i>Packet Delivery</i>	100%	100% of CIR*	99.95%
<b>Availability</b>	<i>Availability, Single Port</i>	99.98%	99.98%	99.98%
	<i>Availability, Dual Port</i>	99.998%	99.998%	99.998%
<b>Jitter</b>	<i>Jitter</i>	2ms	10ms	NA
<b>Bandwidth</b>	<i>Bandwidth</i>	100%	100% of CIR*	NA

# SLA example: Sprintlink

---

	<b>Delay</b>	<b>Packet loss</b>	<b>Availability</b>	<b>Jitter</b>
<b>North America</b>	<b>55 ms</b>	<b>0.30%</b>	<b>99.90%</b>	<b>2 ms</b>
<b>Europe</b>	<b>44 ms</b>	<b>0.30%</b>	<b>99.90%</b>	<b>2 ms</b>
<b>Asia</b>	<b>105 ms</b>	<b>0.30%</b>	<b>99.90%</b>	<b>2 ms</b>
<b>South pacific</b>	<b>70 ms</b>	<b>0.30%</b>	<b>99.90%</b>	<b>2 ms</b>
<b>Continental US (Peerless IP)</b>	<b>55ms</b>	<b>0.1%</b>	<b>n/a</b>	<b>2 ms</b>

# Measurement Technology

---

- **We've known what metrics used to describe network performance, but how to measure them?**
- **Technologies and tools**
  - ping, traceroute,iperf, jperf.
  - **SNMP**
  - **Netflow (Cisco), Sflow (Juniper), NetStream (Huawei)**
  - **IP SLA (Cisco)**
  - **Etc.**

# Active Measurement Tools

---

- **Tools that inject packets into the network to measure some value**
  - Available Bandwidth
  - Delay/Jitter
  - Loss
- **Requires bi-directional traffic or synchronized hosts**

# Passive Measurement Tools

---

- **Tools that monitor existing traffic on the network and extract some information**
  - Bandwidth used
  - Jitter
  - Loss rate
- **May generate some privacy and/or security concerns**

# ping

---

- Normally used as a troubleshooting tool
- Uses ICMP Echo messages to determine:
  - Whether a remote device is active (for trouble shooting)
  - round trip time delay (RTT), **but not one-way delay**
  - Packet loss
- Sometime we need to specify the source and length of packet using extended ping in router or host
  - *Why using large packet when ping?*  
(to test the link quality and throughput.)
  - *Large packet ping is prohibited in Windows, but Linux is ok*

# Sample Ping

---

```
Freebsd>% ping 202.112.60.31
PING 202.112.60.31 (202.112.60.31) 56(84) bytes of data.
64 bytes from 202.112.60.31: icmp_seq=1 ttl=253 time=0.326 ms
.....
64 bytes from 202.112.60.31: icmp_seq=6 ttl=253 time=0.288 ms
6 packets transmitted, 6 received, 0% packet loss, time 4996ms
rtt min/avg/max/mdev = 0.239/0.284/0.326/0.025 ms
```

---

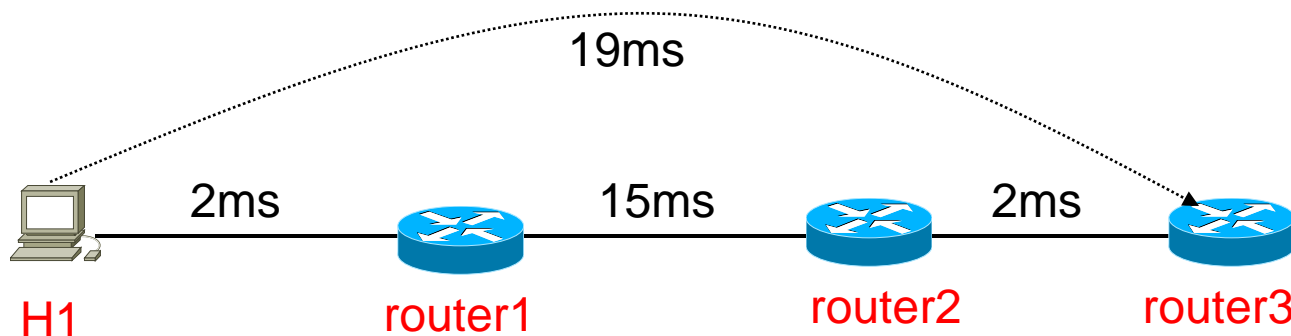
```
router# ping
Protocol [ip]:
Target IP address: 202.112.60.31
Repeat count [5]:
Datagram size [100]: 3000
Timeout in seconds [2]:
Extended commands [n]:
Sweep range of sizes [n]:
Type escape sequence to abort.
Sending 5, 3000-byte ICMP Echos to 202.112.60.31, timeout is 2 seconds:
!!!!
Success rate is 100 percent (5/5), round-trip min/avg/max = 1/1/4 ms
```



# traceroute

---

- Can be used to measure the RTT delay, and also the delay between the routers along the path
- Unix/linux traceroute uses UDP datagram with different TTL to discover the route a packet take to the destination, Microsoft Windows tracert uses ICMP protocol, If Windows tracert appears to show continuous timeouts, the router may be filtering ICMP traffic – try a Unix/Linux traceroute
- After the Nachi worm, many ISPs filter ICMP traffic. So ping can not work, but traceroute is ok



# Sample Traceroute

```
C:\WINDOWS>tracert -d www.163.com

Tracing route to www.cache.split.netease.com [202.108.9.16]
over a maximum of 30 hops:

  0  <1 ms    <1 ms    <1 ms    202.112.0.126
  1  24 ms    14 ms    <1 ms    202.112.53.73
  2  <1 ms    <1 ms    <1 ms    202.112.53.178
  3  <1 ms    <1 ms    <1 ms    202.38.123.18
  4  <1 ms    <1 ms    <1 ms    202.38.123.10
  5  125 ms   124 ms   123 ms   219.158.28.73
  6  125 ms   129 ms   130 ms   219.158.11.121
  7  130 ms   132 ms   132 ms   202.96.12.162
  8  125 ms   123 ms   122 ms   202.106.193.38
  9  122 ms   121 ms   126 ms   202.106.193.166
 10  121 ms   123 ms   122 ms   61.148.3.234
 11  125 ms   127 ms   127 ms   202.108.9.16

Trace complete.
```

```
Router# traceroute 202.112.60.37
```

```
Type escape sequence to abort.
```

```
Tracing the route to 202.112.60.37
```

```
 0 202.112.53.169    0 msec  0 msec  0 msec
 1 202.112.36.250    20 msec 20 msec 16 msec
 2 202.112.36.254    28 msec 28 msec 24 msec
 3 202.112.53.202    24 msec *      24 msec
```

# Visual Route

- Visualization of traceroute information
- <http://www.visualroute.com>

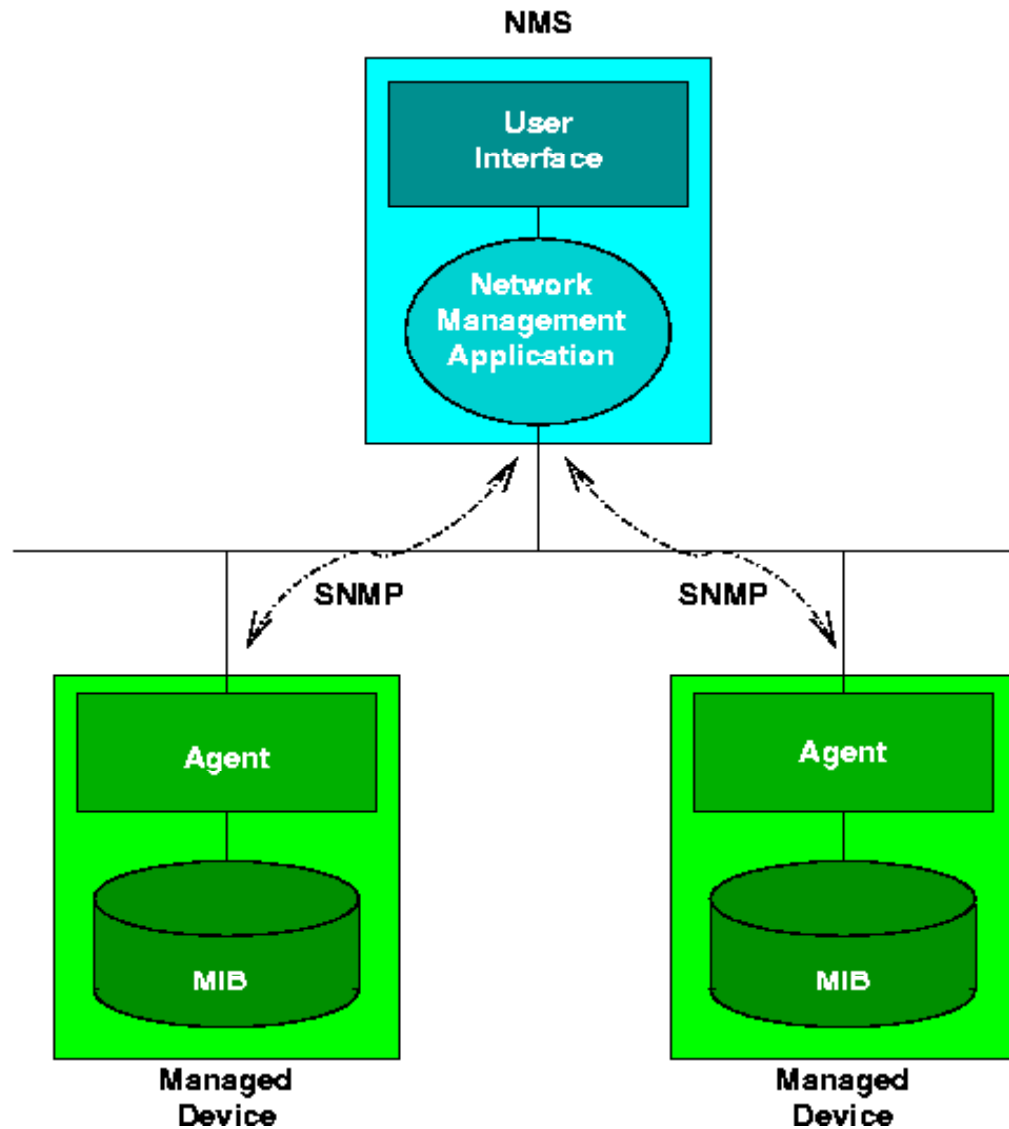
Address:  IP Addresses:

The map displays a route starting from Santa Clara, CA (marked with a question mark) and ending in Beijing, China (marked '1.'). The route passes through Hong Kong (marked '6.').

Hop	%Loss	IP Address	Node Name	Location	Tzone	ms	Graph	Network
0		202.112.228.1	puma	*			0	315 China Education and Resear
1		202.112.228.1	beijing-bgw1.cernet.net	Beijing, China	+08:00	0		China Education and Resear
2		202.112.53.73	cd0.cernet.net	Guangzhou, China	+08:00	0		CERNET super computer cent
3		202.112.53.182	-	Guangzhou, China	+08:00	0		CERNET super computer cent
4		202.112.61.197	-	Beijing, China	+08:00	0		CERNET {mxBgVPPDPEO"2?"}
5		202.112.61.18	-	Beijing, China	+08:00	43		CERNET {mxBgVPPDPEO"2?"}
6		203.222.39.157	sl-gw10-hk-11-0.sprin	Hong Kong, Hong Kong	+08:00	37		Sprintlink Asia Pacific Re
7		203.222.38.37	sl-bb20-hk-14-0.sprin	Hong Kong, Hong Kong	+08:00	37		Sprintlink Asia Pacific Re
8		144.232.9.211	sl-bb24-sj-0-0.sprin	San Jose, CA, USA	-08:00	217		Sprint SPRINT-INNET9
9		144.232.3.249	sl-bb23-sj-15-0.sprin	San Jose, CA, USA	-08:00	224		Sprint SPRINT-INNET9
10		144.232.0.250	sl-gw19-sj-15-0.sprin	San Jose, CA, USA	-08:00	219		Sprint SPRINT-INNET9
11		144.228.111.94	sl-internap-140-0.spr	Seattle, WA, USA	-08:00	224		Sprint SPRINTLINK
12		63.251.63.1	border1.ge1-1-bbnet1.s	Seattle, WA, USA	-08:00	235		InterNAP Network Services,
...								
?		72.5.124.61	www.sun.com	Santa Clara, CA				SUN MICROSYSTEMS INAP-SFO-

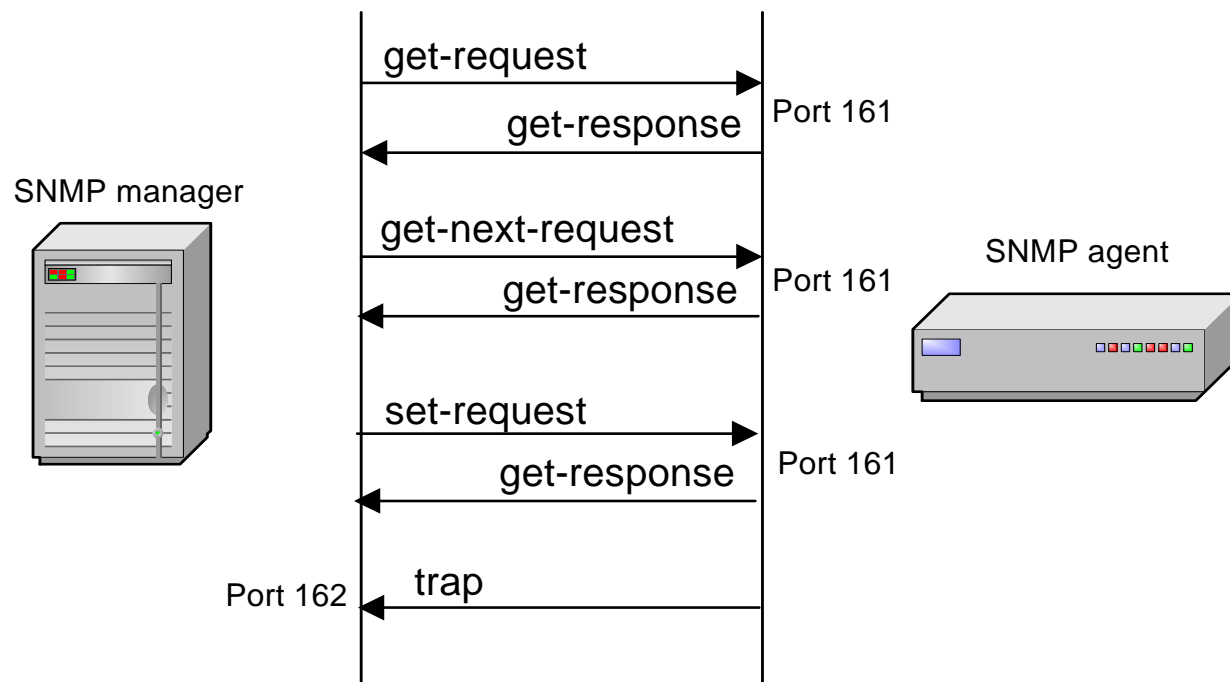
# SNMP Architecture

---



# SNMP Protocol

- C/S based, Client Pull and Server Push
- Ports: UDP 161(snmp messages), UDP 162(trap messages)
- SNMP manager and an SNMP agent communicate using the SNMP protocol
  - Generally: Manager sends queries and agent responds
  - Exception: Traps are initiated by agent.



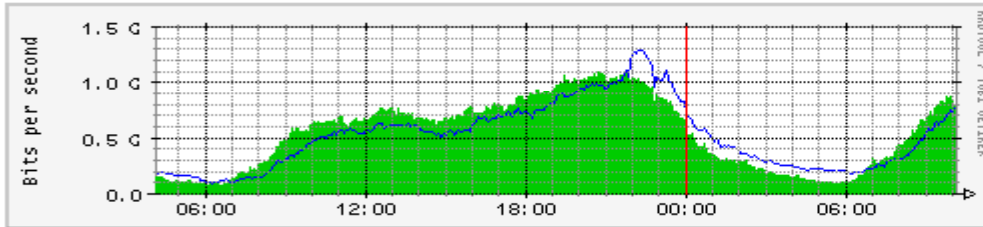
# MRTG

---

- **The Multi Router Traffic Grapher: a freeware written in Perl, works on unix/linux, graph data collected from routers and other devices or applications based on SNMP.**
- **One of most popular network monitoring tools used today: to monitoring the bandwidth utilization of network link**
- **SNMP v2c support, no more counter wrapping**
- **<http://oss.oetiker.ch/mrtg/>**

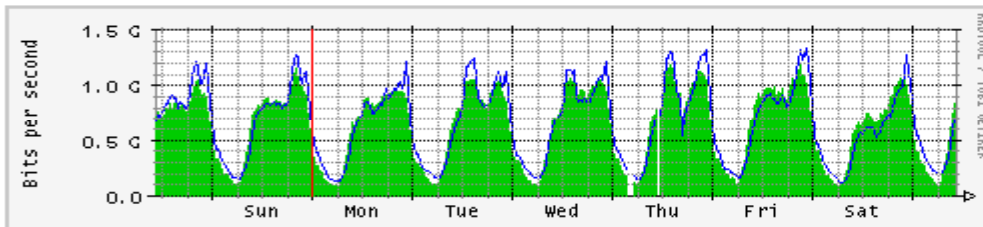
# MRTG Example

## `Daily' Graph (5 Minute Average)



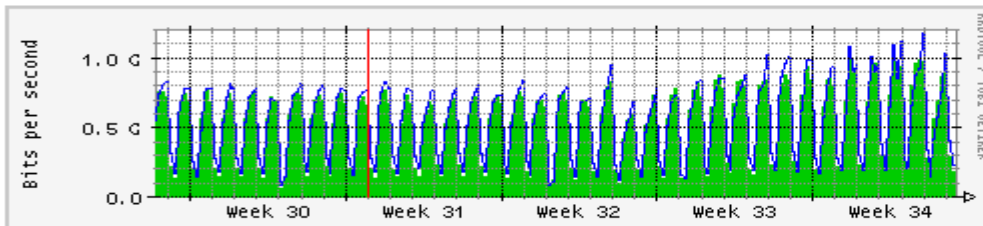
最大 In: 1087.6 Mb/s (43.5%) 平均 In: 537.7 Mb/s (21.5%) 当前 In: 797.4 Mb/s (31.9%)  
最大 Out: 1284.4 Mb/s (51.4%) 平均 Out: 512.9 Mb/s (20.5%) 当前 Out: 764.9 Mb/s (30.6%)

## `Weekly' Graph (30 Minute Average)



最大 In: 1185.3 Mb/s (47.4%) 平均 In: 638.9 Mb/s (25.6%) 当前 In: 841.1 Mb/s (33.6%)  
最大 Out: 1338.4 Mb/s (53.5%) 平均 Out: 656.3 Mb/s (26.3%) 当前 Out: 699.1 Mb/s (28.0%)

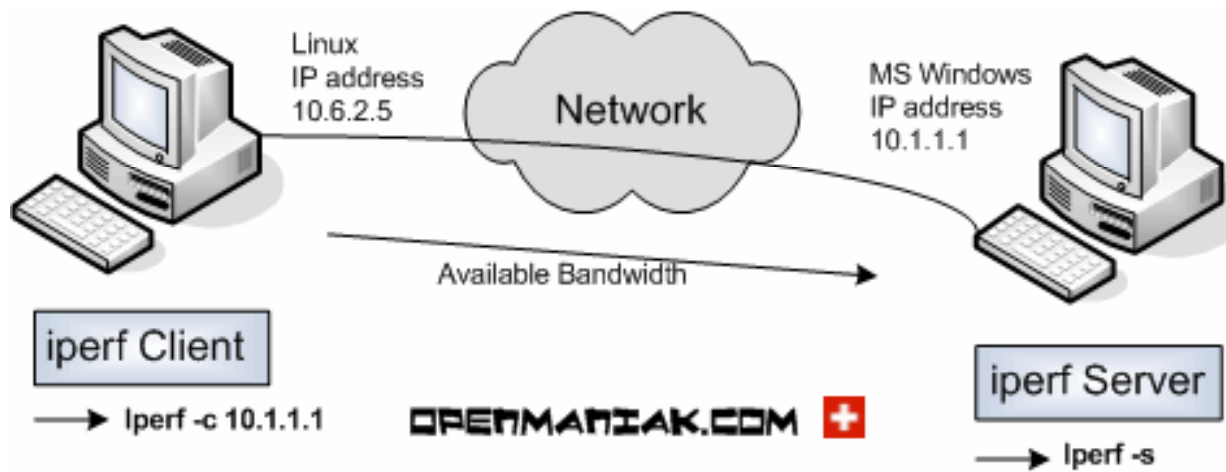
## `Monthly' Graph (2 Hour Average)



最大 In: 1036.8 Mb/s (41.5%) 平均 In: 538.1 Mb/s (21.5%) 当前 In: 674.9 Mb/s (27.0%)  
最大 Out: 1174.0 Mb/s (47.0%) 平均 Out: 560.6 Mb/s (22.4%) 当前 Out: 529.4 Mb/s (21.2%)

# IPerf

- **Client/server application that**
  - Measures maximum TCP performance
  - Facilitates tuning of TCP and UDP parameters
  - Reports bandwidth, jitter, and packet loss
- **<http://dast.nlanr.net/Projects/Iperf/>**





# Contoh iperf

→ Client side:

```
#iperf -c 10.1.1.1 -u -b 10m
```

-----  
Client connecting to 10.1.1.1, UDP port 5001

Sending 1470 byte datagrams

UDP buffer size: 108 KByte (default)  
-----

[ 3] local 10.6.2.5 port 32781 connected with 10.1.1.1 port 5001

[ 3] 0.0-10.0 sec 11.8 MBytes 9.89 Mbits/sec

[ 3] Sent 8409 datagrams

[ 3] Server Report:

[ 3] 0.0-10.0 sec 11.8 MBytes 9.86 Mbits/sec 2.617 ms 9/ 8409 (0.11%)

→ Server side:

```
#iperf -s -u -i 1
```

-----  
Server listening on UDP port 5001

Receiving 1470 byte datagrams

UDP buffer size: 8.00 KByte (default)  
-----

[904] local 10.1.1.1 port 5001 connected with 10.6.2.5 port 32781

[ ID]	Interval	Transfer	Bandwidth	Jitter	Lost/Total Datagrams
-------	----------	----------	-----------	--------	----------------------

[904]	0.0- 1.0 sec	1.17 MBytes	9.84 Mbits/sec	1.830 ms	0/ 837 (0%)
-------	--------------	-------------	----------------	----------	-------------

[904]	1.0- 2.0 sec	1.18 MBytes	9.94 Mbits/sec	1.846 ms	5/ 850 (0.59%)
-------	--------------	-------------	----------------	----------	----------------

[904]	2.0- 3.0 sec	1.19 MBytes	9.98 Mbits/sec	1.802 ms	2/ 851 (0.24%)
-------	--------------	-------------	----------------	----------	----------------

[904]	3.0- 4.0 sec	1.19 MBytes	10.0 Mbits/sec	1.830 ms	0/ 850 (0%)
-------	--------------	-------------	----------------	----------	-------------

[904]	4.0- 5.0 sec	1.19 MBytes	9.98 Mbits/sec	1.846 ms	1/ 850 (0.12%)
-------	--------------	-------------	----------------	----------	----------------

[904]	5.0- 6.0 sec	1.19 MBytes	10.0 Mbits/sec	1.806 ms	0/ 851 (0%)
-------	--------------	-------------	----------------	----------	-------------

[904]	6.0- 7.0 sec	1.06 MBytes	8.87 Mbits/sec	1.803 ms	1/ 755 (0.13%)
-------	--------------	-------------	----------------	----------	----------------

[904]	7.0- 8.0 sec	1.19 MBytes	10.0 Mbits/sec	1.831 ms	0/ 850 (0%)
-------	--------------	-------------	----------------	----------	-------------

[904]	8.0- 9.0 sec	1.19 MBytes	10.0 Mbits/sec	1.841 ms	0/ 850 (0%)
-------	--------------	-------------	----------------	----------	-------------

[904]	9.0-10.0 sec	1.19 MBytes	10.0 Mbits/sec	1.801 ms	0/ 851 (0%)
-------	--------------	-------------	----------------	----------	-------------

[904]	0.0-10.0 sec	11.8 MBytes	9.86 Mbits/sec	2.618 ms	9/ 8409 (0.11%)
-------	--------------	-------------	----------------	----------	-----------------

# Jperf

The screenshot displays the Jperf 2.0 graphical user interface. The window title is ".Jperf 2.0 - Network performance measurement graphical tool".

**Configuration Section:**

- ipperf command:** bin/iperf.exe -s -u -P 0 -l 1 -p 5001 -f m
- Choose iPerf Mode:**  Client,  Server
- Server address:** [empty], **Port:** 5001
- Parallel Streams:** 1
- Listen Port:** 5001
- Client Limit:** [empty]
- Num Connections:** 0

**Application layer options:**

- Enable Compatibility Mode
- Transmit:** 10 [unit: Bytes/Seconds]
- Output Format:** Mbits
- Report Interval:** 1 seconds
- Testing Mode:**  Dual,  Trade
- test port:** 5001
- Print MSS

**Transport layer options:**

- Choose the protocol to use:**  TCP,  UDP
- Buffer Length: 2 Mbytes
- TCP Window Size: 56 Bbytes
- Max Segment Size: 1 Bbytes
- TCP No Delay
- UDP Bandwidth:** 1 Mbytes/sec
- UDP Buffer Size: 41 Bbytes
- UDP Packet Size: 32 Bbytes

**IP layer options:**

- TTL: [empty]

**Bandwidth & Jitter Graph:**

The graph shows performance over 9 seconds. The top chart displays Bandwidth (Mbps) fluctuating around 10.0. The bottom chart displays Jitter (ms) fluctuating around 1.8. Summary statistics are shown below the graph:

- Bandwidth: 10.05 Mbps
- Jitter: 1.87 ms

**Output Log:**

```
bin/iperf.exe -s -u -P 0 -l 1 -p 5001 -f m
Server listening on UDP port 5001
Receiving 1470 byte datagrams
UDP buffer size: 0.01 MByte (default)

[1912] local 192.168.1.2 port 5001 connected with 192.168.1.102 port 32769
[ID] Interval Transfer Bandwidth Jitter Lost/Total Datagrams
[1912] 0.0- 1.0 sec 1.22 Mbytes 10.3 Mbits/sec 1.041 ms 11280/0(0%) 0/0 (1.3e+00%)
[1912] 1.0- 2.0 sec 1.14 Mbytes 9.53 Mbits/sec 1.840 ms 41/ 851 (4.8%)
[1912] 2.0- 3.0 sec 1.13 Mbytes 9.44 Mbits/sec 1.829 ms 48/ 851 (5.6%)
[1912] 3.0- 4.0 sec 1.13 Mbytes 9.48 Mbits/sec 1.041 ms 45/ 851 (5.3%)
[1912] 4.0- 5.0 sec 1.15 Mbytes 9.52 Mbits/sec 1.146 ms 33/ 841 (3.9%)
```

# Performance Management Process

---

