

Review from socket programming chapters:

UDP Socket Functions:

- socket(), bind(), recvfrom(), sendto(), close()

TCP Socket Functions:

- socket(), bind(), listen(), accept(), connect(), recv(), send(), read(), write(), close()

Other commonly used ones:

- setsockopt(), gethostbyname(), htons(), htonl(), ntohs(), ntohl(), bzero(), etc.

Debugging:

- return value, localhost, nc, tcpdump

Network Performance

- Bandwidth (aka throughput)

* A **bandwidth** of a network is the # of bits that can be transmitted over the network in a period of time

$$\text{Throughput} = \text{TransferSize} / \text{TransferTime}$$

* 100 Mbps => transfer 100 M bits per second

* can be defined for a single link or end-end channel

- Bandwidth vs Throughput

* Literally, bandwidth is a measure of **width of frequency band**. It is measured in *hertz*

* When we talk about bandwidth of a communication link, it measured in **bits per second** (data rate)

* **Throughput** is often used to refer to the measured performance of a system

* Due to efficiency problems, throughput may be smaller than the bandwidth of a link (e.g., a link with 10 Mbps may achieve a throughput of 2 Mbps)

- Latency (aka delay)

* Describes how long it takes a message to travel across the network; measured strictly in terms of time

* can be defined for a single link or end-end channel

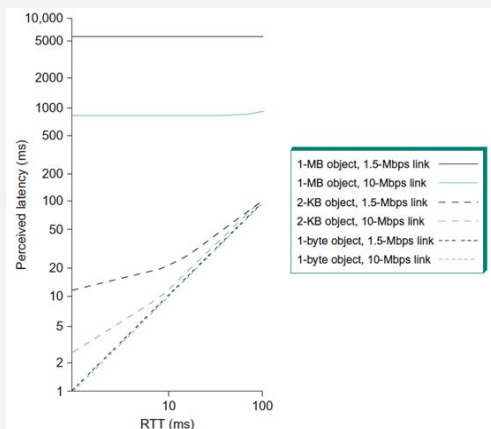
* **Round Trip Time (RTT)** is more important sometimes

* Components of Network Latency:

$$\text{Latency} = \text{Propagation} + \text{Transmit} + \text{Queue}$$

$$\text{Propagation} = \text{Distance} / \text{Speed of Light}$$

$$\text{Transmit} = \text{Size} / \text{Bandwidth}$$



<= when object size is large, transmission delay dominates

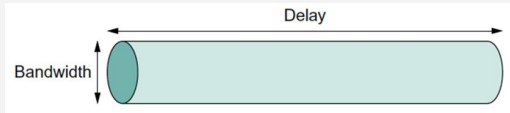
<= when object size is medium, both transmission & propagation matter

<= when object size is small, propagation delay dominates

- Bandwidth & Latency define the performance of a link/channel

- * Latency dominates some apps (e.g., a client sends 1 byte to the server)
- * Bandwidth dominates other apps (e.g., fetch a 1 GB file from a FTP server)
- * Who dominates depends on the size of the data, the bandwidth of the link/channel, the latency of the link/channel

- Delay * Bandwidth...



* Think of channel as a hollow pipe; latency = length, and bandwidth = diameter of pipe

* *The product of the 2 gives the volume of the pipe:*

max # of bits that could be in transit through the pipe at a given instant

* For example, a channel with one-way latency of 100 ms and a bandwidth of 5 Mbps is able to hold:

$$100 \cdot 10^{-3} \text{ s} \cdot 5 \cdot 10^6 \text{ bits/s} = 500 \text{ K bits} = 62.5 \text{ KB}$$

* If the app considers the round-trip delay, the product is $\text{RTT} \cdot \text{Bandwidth}$

$$200 \cdot 10^{-3} \text{ s} \cdot 5 \cdot 10^6 \text{ bits/s} = 1 \text{ M bits} = 125 \text{ KB}$$

* It corresponds to the number of bits a sender can transit before hearing something back from receiver (most of the time, we use RTT in the delay * Bandwidth product)

(some rough RTT values for quick-dirty calculation: 100 ms for cross-country RTT; 1 ms for a local network)

Example: Transfer a 1 MB file over a 1 Mbps network vs a 1 Gbps network. RTT for both network is 100 ms.

Q: What will be the transfer time and throughput of the two networks?

$$\text{TransferTime} = \text{RTT} + \text{TransferSize} / \text{Bandwidth}$$

$$\text{Throughput} = \text{TransferSize} / \text{TransferTime}$$

For 1 Mbps:

$$\text{Transfer time: } 100 \text{ ms} + 1 \text{ MB} / 1 \text{ Mbps} = 8.1 \text{ s}$$

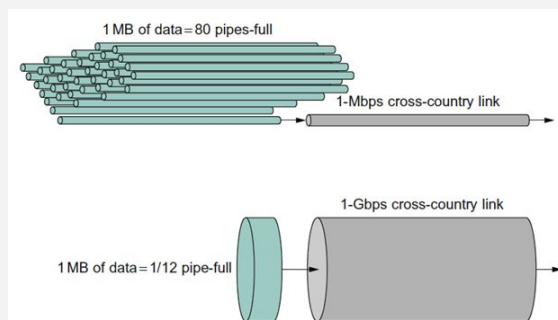
$$\text{Throughput: } 1 \text{ MB} / 8.1 \text{ s} \approx 0.99 \text{ Mbps}$$

For 1 Gbps:

$$\text{Transfer time: } 100 \text{ ms} + 1 \text{ MB} / 1 \text{ Gbps} = 108 \text{ ms}$$

$$\text{Throughput: } 1 \text{ MB} / 108 \text{ ms} \approx 74.1 \text{ Mbps}$$

Because of RRT, throughput of 1 Gbps is only 74 times (not 1000 times) higher than the throughput of 1 Mbps!



\Leftarrow 1 MB of data is sent like a **stream** on the 1Mbps channel

\Leftarrow 1 MB of data is sent like a **packet** on the 1Gbps channel

- Network Jitter

- * Describes how much the **latency varies** from packet to packet
- * Very important for time-sensitive apps such as streaming and video applications
- * Understanding the lower and upper bounds of packet latency would help reduce jitter (e.g., app can delay the play of the first frame)



What's next on Computer Networks?

- Ubiquitous networking
 - Get internet access anywhere & anytime (e.g. planes and trains, etc.)
 - Interconnect a great variety of devices (computers, mobile phones, sensors, etc.)
- Scalability
 - Support several orders of magnitude more devices
 - Connect to large datacenters filled with tens of thousands processors and many petabytes of data (Cloud Computing)
- Applications
 - Smart Home, Smart City, Smart Planet
 - Internet of Things