# A Quick Review of Computer Networking Architecture, Applications, Transport (TCP), Routing

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February 19, 2020

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■ *Packet switch:* a *link-layer switch* or a *router* 

- Communication link: a connection between packet switches and/or end systems
- **Route:** sequence of switches that a packet goes through (a.k.a. *path*)
- Protocol: control the sending and receiving of information to and from end systems and packet switches

## **Communication Links**

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- Various types and forms of medium
  - Fiber-optic cable
  - Twisted-pair copper wire
  - Coaxial cable
  - Wireless local-area links (e.g., 802.11, Bluetooth)
  - Satellite channel
  - ▶ ...







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- A switch (router) receives packets and *forwards* them along to other switches or to end systems
- Every forwarding decision is taken on the basis of the information contained in the packet





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- Communication requires a connection setup phase in which the network reserves all the necessary resources for that connection (links, buffers, switches, etc.)
- After a successful setup, the communicating systems are connected by *a set of links dedicated to the connection* for the entire duration of their conversation
- When the conversation ends, the network tears down the connection, freeing the corresponding resources (links, buffers, etc.) for other connections

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- Circuit switching requires an expensive setup phase
  - however, once the connection is established, little or no processing is required
- Packet switching does not incur any setup cost
  - however, it always incurs a significant processing and space overhead, on a per-packet basis
    - processing cost for forwarding
    - space overhead because every packet must be self-contained

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- Circuit switching admits a straightforward implementation of quality-of-service guarantees
  - network resources are reserved at connection setup time
- Guaranteeing any quality of service with packet switching is very difficult
  - no concept of a "connection"
  - and again, processing, space overhead, etc.

## **Circuit vs. Packet Switching (3)**

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  - i.e., circuit switching is an inefficient way to use the network

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  - i.e., circuit switching is an inefficient way to use the network
- Packet switching achieves a much better utilization of network resources
  - it is designed specifically to share links
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## Virtual Circuits

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- Information is sent in packets, so links can be shared more effectively
- Packets carry a *virtual circuit identifier* instead of the destination address
  - Important observation: at any given time there are much fewer connections than destinations
    - much faster per-packet processing (forwarding)
    - Iower per-packet space overhead







communication network









# **Service Perspective**



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■ What kind of *service* does the Internet offer to end systems?

# **Type of Service**

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- the network accepts "datagrams" for delivery—this is conceptually similar to the postal service
- "best effort" really means unreliable though not malicious

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- "best effort" really means unreliable though not malicious

#### Connection-oriented, reliable

- ► virtual duplex communication channel (A ↔ B)—conceptually similar to a telephone service
- information is transmitted "reliably" and in order

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- How reliable is a "reliable" service?
- The term "reliable" means that information will eventually reach its destination if a route is viable within a certain amount of time
- The network makes absolutely no guarantees on *latency* (i.e., the time it takes to transmit some information from a source to a destination)

application

application

transport

application transport network

application transport network link

application transport network link physical

- application functionalities
- application messages

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#### ■ Application (e.g., HTTP, SMTP, and DNS)

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#### Physical














Propagation <b>Delay</b>	$d_{prop} = t_1 - t_0$	sec
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$$d_{end-end} = d_1 + \frac{\ell}{R_1} + d_x + \frac{\ell}{R_2} + d_2$$









 $R_{end-end} =$ 



 $R_{end-end} = \min\{R_1, R_2\}$ 



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 $R_{end-end} = \min\{R_1, R_2, \ldots, R_N\}$ 









### **Queuing Delay**



where

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 $\ldots R_x$  is also the rate at which packets get out of the queue

HTTP




































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- Terminology
  - transport-layer packets are called segments
- Basic assumptions on the underlying network layer
  - every host has one unique IP address
  - best-effort delivery service
    - no guarantees on the integrity of segments
    - no guarantees on the order in which segments are delivered

#### Transport-layer multiplexing/demultiplexing

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#### Congestion control

 i.e., end-to-end traffic (admission) control so as to avoid destructive congestions within the network











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- How do we find out which application (host and port number) to connect to?
  - outside the scope of the definition of the transport layer
  - but of course we can have "well-known" service numbers

# **Ports**
The message format of both UDP and TCP starts with the source and destination port numbers

0	1!	516 31
	source port	destination port

•••

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E.g.,

0	15 <sup>-</sup>	16 31
source port		destination port



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- Full-duplex service
  - both endpoints can both send and receive, at the same time

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Maximum transmission unit (MTU): largest link-layer frame available to the sender host

path MTU: largest link-layer frame that can be sent on all links from the sender host to the receiver host

# **TCP Segment Format**

0 31						
source port					destination port	
sequence number						
acknowledgment number						
hdrlen	unused	U	A P	R S	F	receive window
Internet checksum					urgent data pointer	
options field						
data						

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- *Checksum:* (16-bit) used to detect transmission errors

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$$[Seq\# = 1200, ...], size(data) = 1000 \longrightarrow$$
  
 $[Seq\# = 2200, ...], size(data) = 500 \longrightarrow$   
 $[Seq\# = ..., Ack\# = 2700]$ 

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$$[Seq\# = 201, Ack\# = 102, Data = "i"] \longrightarrow$$

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Acknowledgments are "piggybacked" on data segments

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- **Retransmission timeouts should be larger than the round-trip time** RTT = 2L
  - as close as possible to the RTT
- TCP controls its timeout by continuously *estimating the current RTT*

- RTT is measured using ACKs
  - only for packets transmitted once
- Given a single sample *S* at any given time

Exponential weighted moving average (EWMA)

$$\overline{RTT} = (1 - \alpha)\overline{RTT}' + \alpha S$$

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$$\alpha = 0.125$$

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■ TCP also measures the *variability of RTT* 

$$\overline{DevRTT} = (1 - \beta)\overline{DevRTT}' + \beta|\overline{RTT}' - S|$$

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- However, *T* should not be too far from RTT
  - so as to detect (and retransmit) lost segments as quickly as possible
- TCP sets its timeouts using the estimated RTT ( $\overline{RTT}$ ) and the variability estimate  $\overline{DevRTT}$ :

$$T = \overline{RTT} + 4\overline{DevRTT}$$

#### **Reliable Data Transfer (Sender)**

#### A simplified TCP sender

Image r\_send(data)
 if (timer not running)
 start\_timer()
 u\_send([data,next\_seq\_num])
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#### timeout

u\_send(pending segment with smallest sequence number) start\_timer()

### **Reliable Data Transfer (Sender)**

#### A simplified TCP sender

r\_send(data)
if (timer not running)
start\_timer()
u\_send([data,next\_seq\_num])
next\_seq\_num ← next\_seq\_num + length(data)

#### timeout

u\_send(pending segment with smallest sequence number) start\_timer()

u\_recv([ACK,y])
if (y > base)
base ← y
if (there are pending segments)
start\_timer()
else...

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#### else

```
ack\_counter[y] \leftarrow ack\_counter[y] + 1
if (ack\_counter[y] = 3)
u_send(segment with sequence number y)
```

Three-way handshake

Three-way handshake

client server

Three-way handshake



#### Three-way handshake

#### Three-way handshake

"This is it." "Okay, Bye now." "Bye."













"This is it." "Okay, Bye now." "Bye."














































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#### **Extreme case:** constant input data rate

$$\lambda_{in} > R$$

In this case  $|q| = (\lambda_{in} - R)t$  and therefore

$$d_q = \frac{\lambda_{in} - R}{R}t$$

■ Steady-state queuing delay

$$d_q = \begin{cases} 0 & \lambda_{in} < R\\ \frac{\lambda_{in} - R}{R}t & \lambda_{in} > R \end{cases}$$

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What to do when the network is congested?













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- 2. how does the sender *set its output rate*?
  - we need accelerator and brakes to speed up or slow down
- 3. how should the sender *control its output rate*?
  - we need a brain and we need to know how to drive!

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- Congestion means that some queues overflow in one or more routers between the sender and the receiver
  - the visible effect is that some segments are dropped
- Therefore the sender assumes that the network is congested when it (the sender) detects a segment loss
  - duplicate acknowledgements (i.e., NACK)
  - time out (i.e., no ACKs at all)

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The resulting maximum output rate is roughly

$$\lambda = \frac{W}{2L}$$

## **Congestion Control (Brain, Algorithm)**
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- Reaction to timeout events

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e.g., suppose W = 14600 and MSS = 1460, then the sender increases W to 16060 after 10 acknowledgments acknowledgments

■ Window size *W* over time



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- This process is called "slow start" because of the small initial value of W

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- So, TCP reacts differently to a timeout and to a triple duplicate ACKs

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■ NACK (i.e., triple duplicate-ack)

- set ssthresh =  $\overline{W}/2$
- cut W in half:  $W = \overline{W}/2$
- run congestion avoidance, ramping up W linearly
- This is called *fast recovery*

































# Datagram Network
































■ Potentially *multiple paths* for the same source/destination



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## Forwarding

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- Output: output port
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Issues

- how big is the forwarding table?
- how fast does the router have to forward datagrams?
- how does the router build and maintain the forwarding table?

# Routing

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# Routing











Finding paths through a network

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• Example:  $a \rightarrow j$ ?

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- A *cost* function  $c : E \to \mathbb{R}$ 
  - costs are always positive: c(e) > 0 for all  $e \in E$
  - ► links are symmetric: c(u, v) = c(v, u) for all  $u, v \in N$

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■ Compile *u*'s forwarding table by adding the following entry:

$$A(v) \to I_u(x_1)$$

- A(v) is the address (or set of addresses) of router v
- $I_u(x_1)$  is the interface that connects u to the first next-hop router  $x_1$  in  $P_{u \to v} = u, x_1, x_2, \dots, x_n, v$

### **Back To The Example**





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- least-cost path is  $P_{a \rightarrow j} = a, e, b, f, j$
- *a*'s forwarding table will contain an entry  $j \rightarrow 2$  since  $I_a(e) = 2$

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- Link-state routing

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- Every router sends its LSA to every other router in the network, so we need a broadcast routing scheme
- Once we have all the LSAs from every router, and therefore we complete knowledge of G, we need an *algorithm to compute least-cost paths in a graph*

### **Flooding**

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  - cycles in the network create packet storms

- every router forwards a broadcast packet to every adjacent router, except the one where it received the packet router
- a router u accepts a broadcast packet p originating at router s only if p arrives on the link that is on the direct (unicast) path from u to s

#### Reverse-path broadcast

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- Any problem with this solution?
  - it requires (unicast) routing information
  - so it is obviously useless to implement a routing algorithm

### Sequence-number controlled flooding

• the originator *s* of a broadcast packet marks the packet with a sequence number *n*<sub>s</sub>

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- *u* updates its table of sequence numbers  $n_s \leftarrow seq(p)$

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- Gateway routers connect an autonomous system with other autonomous systems
- An *intra-autonomous system routing protocol* runs within an autonomous system (e.g., OSPF)
  - this protocol determines internal routes
    - internal router  $\leftrightarrow$  internal router
    - ▶ internal router ↔ gateway router
    - ▶ gateway router ↔ gateway router







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Both inter-AS and intra-AS routing information is used to compile the forwarding tables

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  - what if *x* is reachable through multiple gateway routers  $G_x, G'_x, \ldots$ ?
    - use *intra-AS* routing information to determine the costs of the (least-cost) paths to  $G_X, G'_X, \ldots$
    - "hot-potato" routing: send it through the closest gateway

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- External subnet addresses are likely to "aggregate" in groups that admit compact representations
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- **BGP** external session (eBGP): a session across two autonomous systems
- **BGP** internal session (iBGP): a session within an autonomous system
  - note that internal sessions carry *inter-AS* information
  - intra-AS routing uses a separate protocol (e.g., OSPF)

# Gateway Routers and *eBGP*



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BGP import policy: used to decide whether to accept or reject the route advertisement

 e.g., a router may not want to send its traffic through one of the AS listed in AS-PATH

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