A Quick Review of Computer Networking

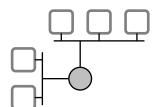
Architecture, Applications, Transport (TCP), Routing

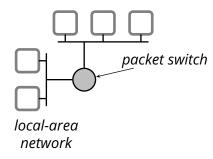
Antonio Carzaniga

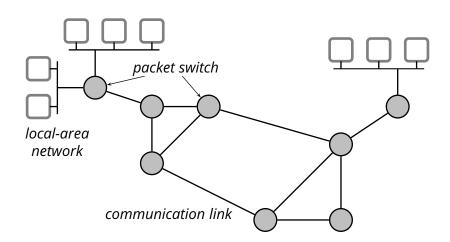
Faculty of Informatics Università della Svizzera italiana

February 22, 2021









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- *Route:* sequence of switches that a packet goes through (a.k.a. *path*)

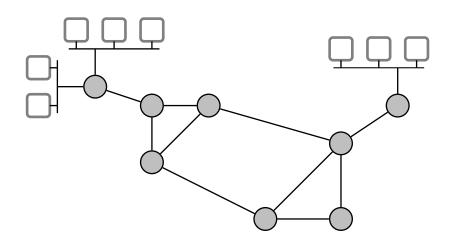
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- **Packet switch:** a link-layer switch or a **router**
- Communication link: a connection between packet switches and/or end systems
- *Route:* sequence of switches that a packet goes through (a.k.a. *path*)
- **Protocol:** control the sending and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

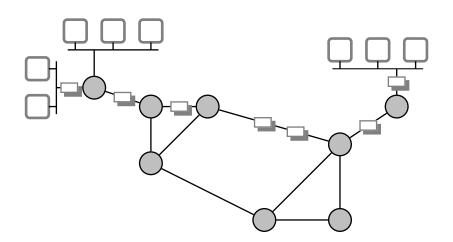
Communication Links

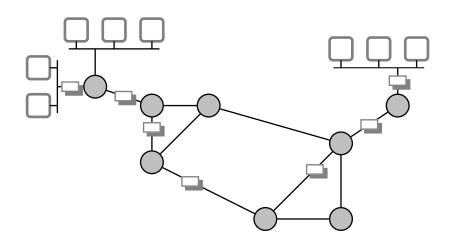
■ Various types and forms of medium

Communication Links

- 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
 - **.**..







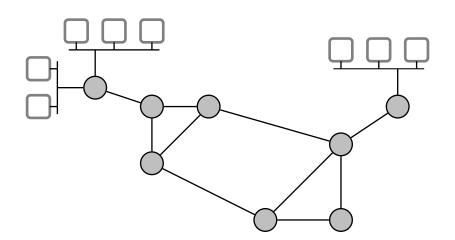
■ The Internet is a *packet-switched* network

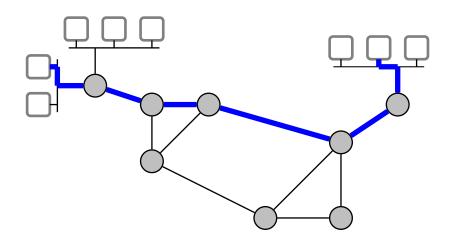
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- Information is transmitted in *packets*
- Switches operate on individual packets
- A switch (router) receives packets and *forwards* them along to other switches or to end systems
- Every forwarding decision is taken on the basis of the information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



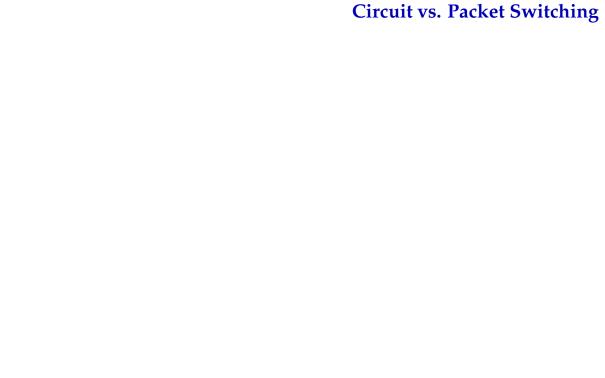


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- Communication requires a connection setup phase in which the network reserves all the necessary resources for that connection (links, buffers, switches, etc.)
- After a successful setup, the communicating systems are connected by *a set of links 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

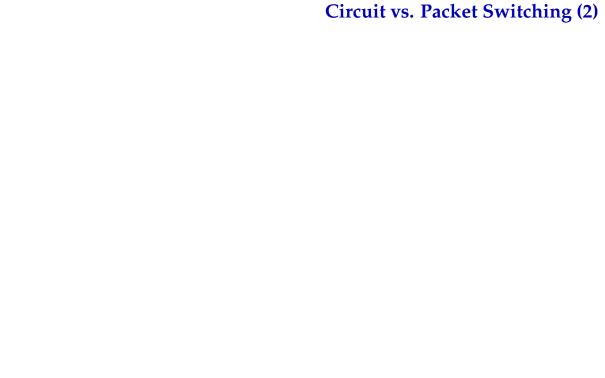


Circuit vs. Packet Switching

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Circuit vs. Packet Switching

- 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



Circuit vs. Packet Switching (2)

- Circuit switching admits a straightforward implementation of quality-of-service guarantees
 - network resources are reserved at connection setup time

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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)

- Circuit switching allows only a limited sharing of communication resources
 - once a connection is established, the resources are blocked even though there might be long silence periods
 - i.e., circuit switching is an inefficient way to use the network

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- Circuit switching allows only a limited sharing of communication resources
 - once a connection is established, the resources are blocked even though there might be long silence periods
 - 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

■ Idea: combine the advantages of circuit switching and packet switching

Virtual Circuits

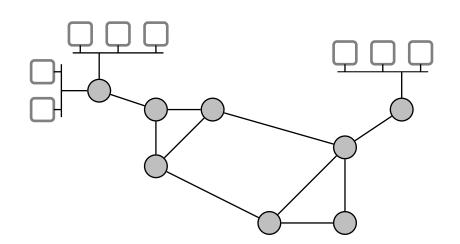
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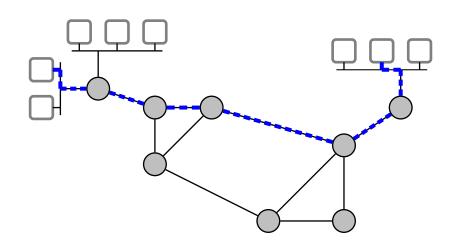
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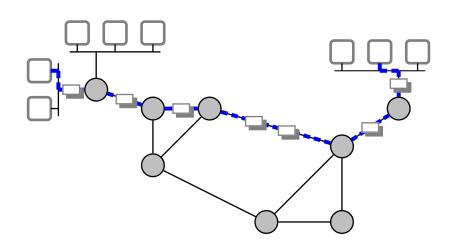
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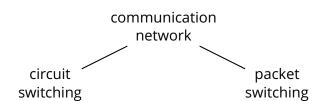
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- The connection does not create a physical circuit, but rather a "virtual circuit"
- Information is sent in packets, so links can be shared more effectively
- Packets carry a *virtual circuit identifier* instead of the destination address
 - Important observation: at any given time there are much fewer connections than destinations
 - much faster per-packet processing (forwarding)
 - lower per-packet space overhead

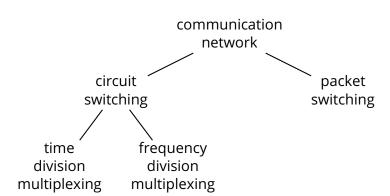


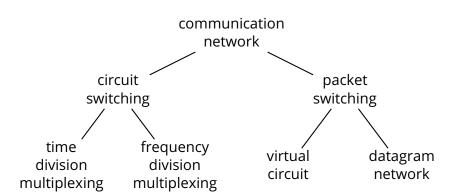


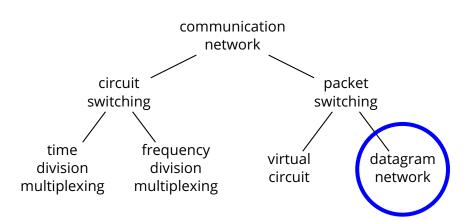


communication network

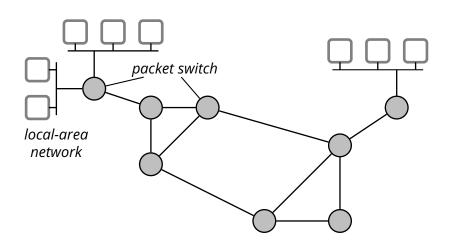




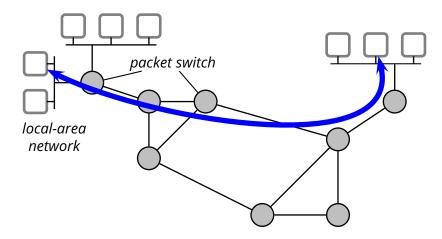




Service Perspective



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

Type of Service

■ Two end systems can communicate through the Internet, but exactly what kind of communication service is that of the Internet?

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■ Connection-oriented, reliable

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

Type of Service (2)

■ How reliable is a "reliable" service?

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- The term "reliable" means that information will eventually reach its destination if a route is viable within a certain amount of time

Type of Service (2)

- 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	

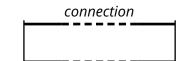
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 - application functionalities
 - application messages

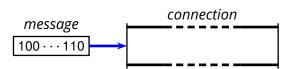
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 - application multiplexing, reliable transfer (TCP), congestion control (TCP)
 - datagrams (UDP) or segments (TCP)

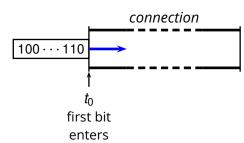
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 - end to end datagram, best-effort service, routing, fragmentation
 - packets (IP)

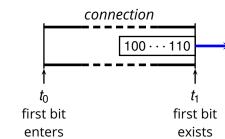
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 - point-to-point or local broadcast communication
 - frames (or packets)

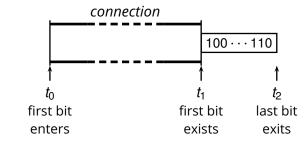
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- Physical

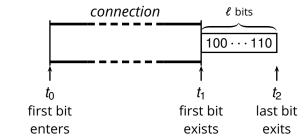


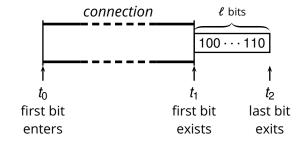




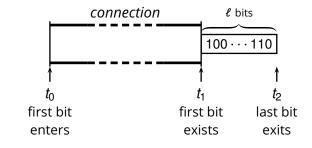








Propagation **Delay** $d_{prop} = t_1 - t_0$ sec



 $d_{prop} = t_1 - t_0$

Propagation **Delay**

$$t_0$$
 t_1 t_2 first bit last bit enters exists exits

Propagation **Delay**
$$d_{prop} = t_1 - t_0$$
 sec

Transmission **Rate**
$$R = \frac{\ell}{t_2 - t_1}$$
 bits/sec

Total transfer time
$$d_{end-end} = d + \frac{\ell}{R}$$
 sec



 H_1



 H_2

$$H_1$$
 d_1, R_1 d_2, R_2 H_2

$$\begin{array}{c|cccc}
d_1, R_1 & d_x & d_2, R_2 \\
\hline
H_1 & X & H_2
\end{array}$$

$$H_1$$
 d_1, R_1 d_2, R_2 H_2

$$d_{end\text{-}end} = d_1 + \frac{\ell}{R_1}$$

$$H_1$$
 d_1, R_1 d_2, R_2 H_2

$$d_{end\text{-}end} = d_1 + \frac{\ell}{R_1} + d_x$$

$$H_1$$
 d_1, R_1 d_2, R_2 H_2

$$d_{end\text{-}end} = d_1 + \frac{\ell}{R_1} + d_x + \frac{\ell}{R_2}$$

$$d_1, R_1$$
 d_2, R_2

$$d_{end\text{-}end} = d_1 + \frac{\ell}{R_1} + d_x + \frac{\ell}{R_2} + d_2$$

$$d_1, R_1$$
 d_2, R_2

$$d_{end\text{-}end} = d_1 + \frac{\ell}{R_1} + d_x + \frac{\ell}{R_2} + d_2$$

$$H_1 \qquad H_2$$

$$d_p, R \qquad d_x \qquad d_p, R \qquad d_x \qquad d_p, R \qquad X_3 \qquad \dots \qquad X_4$$

$$H_1$$
 d_1, R_1 d_2, R_2

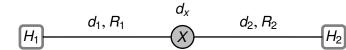
$$d_{end-end} = d_1 + \frac{\ell}{R_1} + d_x + \frac{\ell}{R_2} + d_2$$

$$d_{end-end} = d_1 + \frac{1}{R_1} + d_x + \frac{1}{R_2} + d_2$$

$$d_p, R \xrightarrow{d_x} d_p, R \xrightarrow{d_x} d_p, R \xrightarrow{d_x} \underbrace{d_p, R} \underbrace{X_3} \cdots \underbrace{X_n}$$

$$d_{end\text{-}end} = N\left(d_p + \frac{\ell}{R} + d_x\right)$$





$$d_1, R_1$$
 d_2, R_2 d_2 d_3

$$R_{end-end} =$$

$$d_1, R_1$$
 d_2, R_2 d_2 d_3

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

$$H_1$$
 d_1, R_1 d_2, R_2 H_2

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

$$H_1 \xrightarrow{d_p, R_1} \xrightarrow{d_x} \xrightarrow{d_p, R_2} \xrightarrow{d_x} \xrightarrow{d_p, R_3} \xrightarrow{d_x} \xrightarrow{X_3} \dots$$

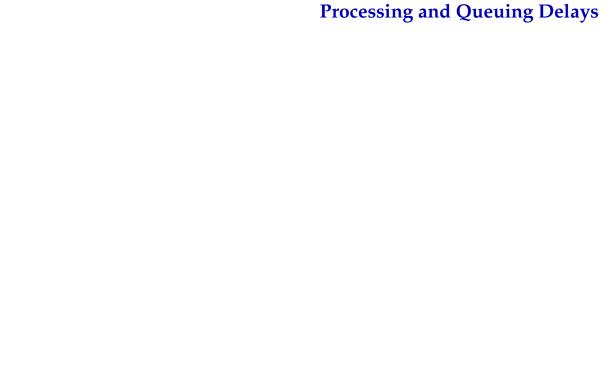
$$\begin{array}{ccc}
d_1, R_1 & d_2, R_2 \\
\hline
H_1 & X & d_2, R_2
\end{array}$$

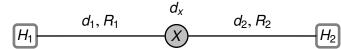
$$\begin{array}{ccc}
H_2 & H_2
\end{array}$$

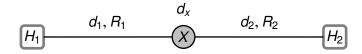
$$R_{end\text{-}end} = \min\{R_1, R_2\}$$

$$H_1 \xrightarrow{d_p, R_1} \xrightarrow{d_x} \xrightarrow{d_p, R_2} \xrightarrow{d_x} \xrightarrow{d_p, R_3} \xrightarrow{d_x} \underbrace{d_x} \xrightarrow{d_p, R_3} \underbrace{d_x} \xrightarrow{d_x} \underbrace{d_x} \underbrace{d_x} \underbrace{d_x} \xrightarrow{d_x} \underbrace{d_x} \underbrace{d_$$

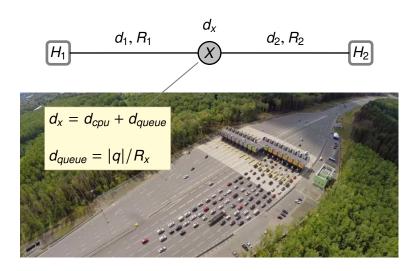
$$R_{end-end} = \min\{R_1, R_2, \dots, R_N\}$$

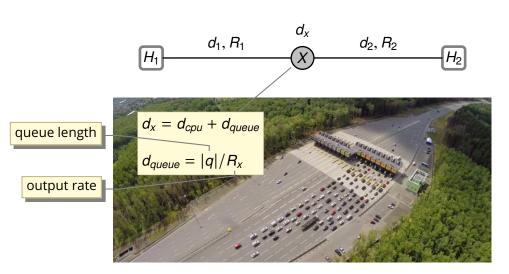






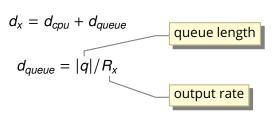






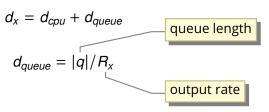
Queuing Delay

where



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where



 $\dots R_X$ is also the rate at which packets get out of the queue

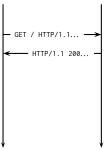


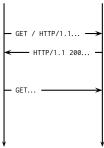
HTTP

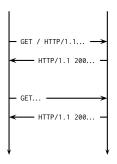
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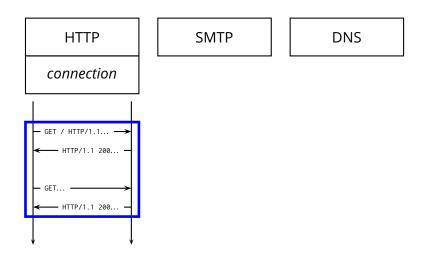
SMTP

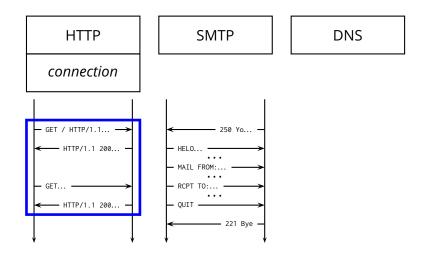
HTTP	SMTP	DNS

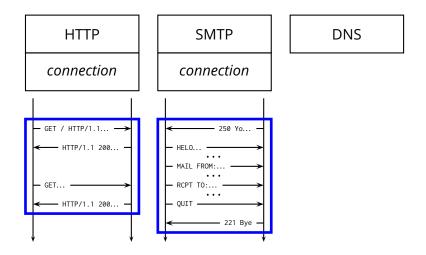


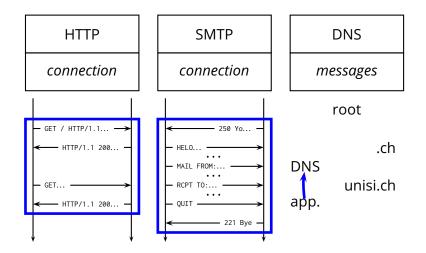


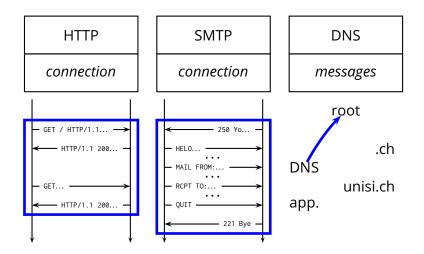


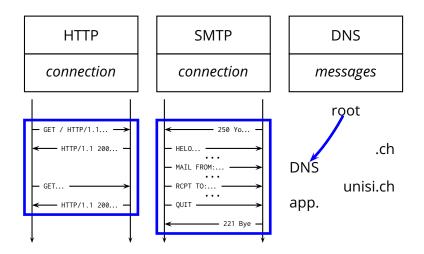


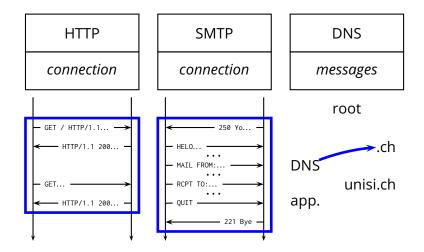


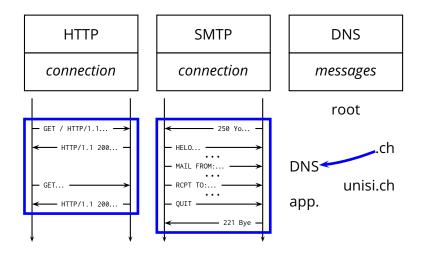


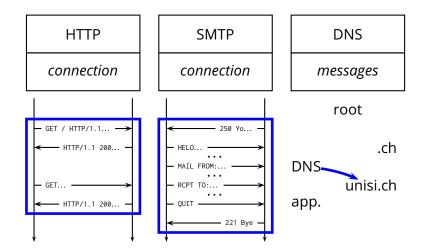


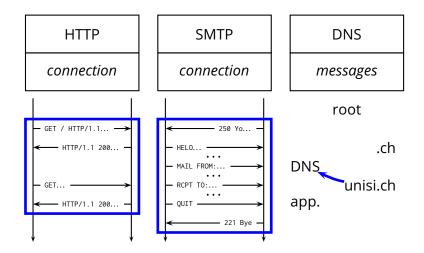


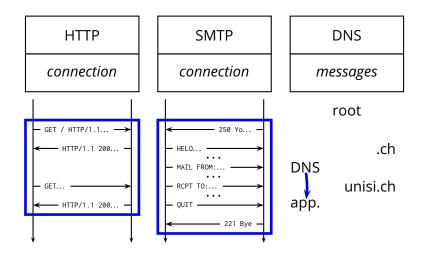


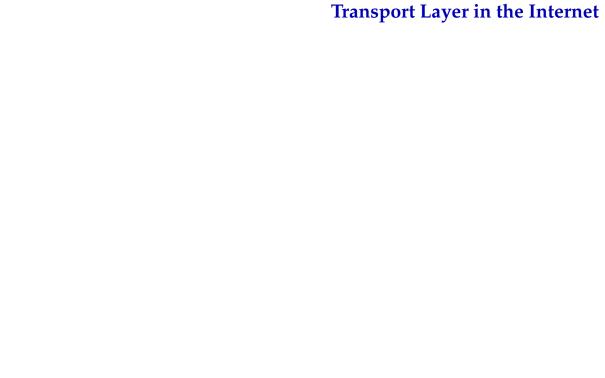












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 - conntection-oriented (i.e., "connections")

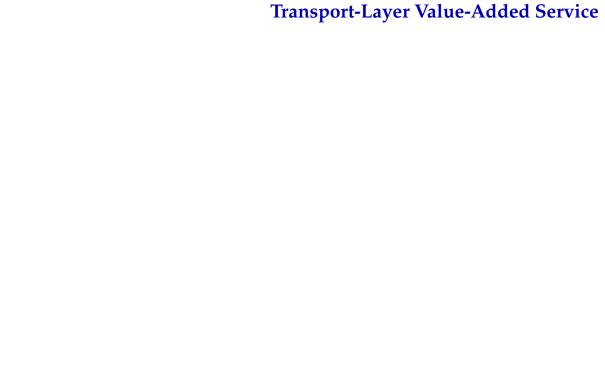
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- Terminology
 - transport-layer packets are called **segments**
- Basic assumptions on the underlying network layer
 - every host has one unique IP address
 - best-effort delivery service
 - no guarantees on the integrity of segments
 - no guarantees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



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 - ▶ i.e., connecting applications as opposed to hosts

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Connections

- i.e., streams
- can be seen as the same as ordered delivery

■ Transport-layer multiplexing/demultiplexing

▶ i.e., connecting applications as opposed to hosts

■ Reliable data transfer

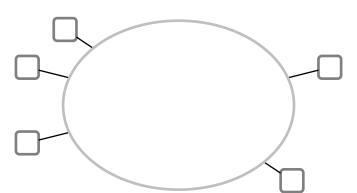
i.e., integrity and possibly ordered delivery

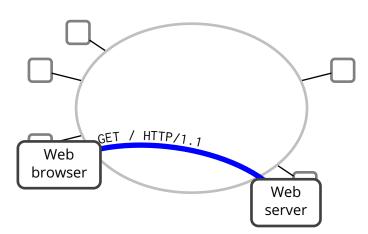
Connections

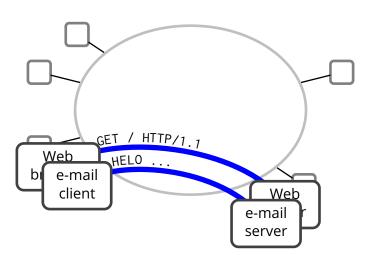
- i.e., streams
- can be seen as the same as ordered delivery

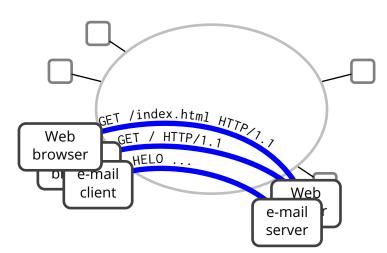
■ Congestion control

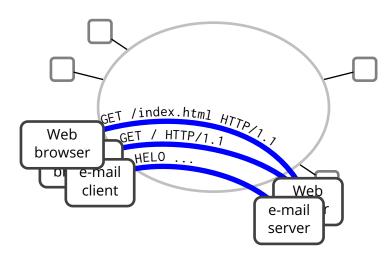
• i.e., end-to-end traffic (admission) control so as to avoid destructive congestions within the network



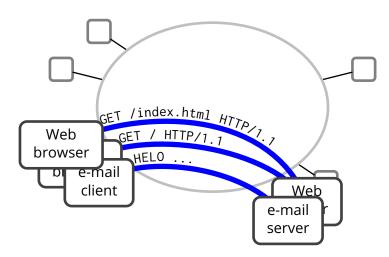








How do we distinguish all these "connections"?



How do we distinguish all these "connections"? (in this case, connections between the same two hosts)



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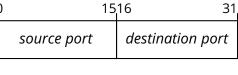
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- How do we find out which application (host and port number) to connect to?
 - outside the scope of the definition of the transport layer
 - but of course we can have "well-known" service numbers

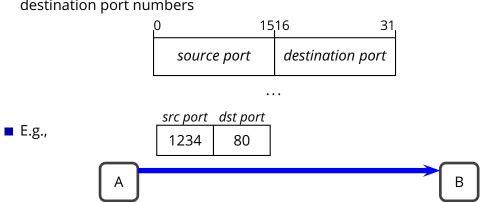


■ The message format of both UDP and TCP starts with the source and destination port numbers

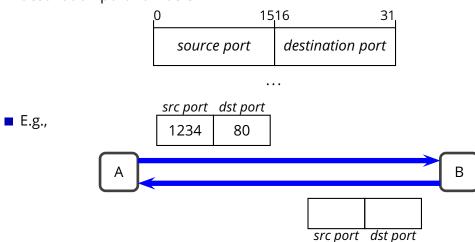


• •

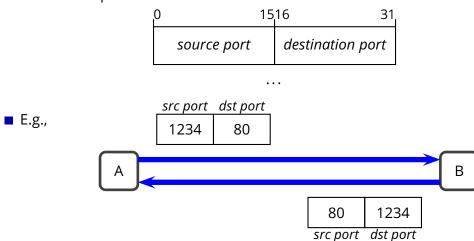
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- Full-duplex service
 - both endpoints can both send and receive, at the same time



Preliminary Definitions

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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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- Optional and variable-length options field: may be used to negotiate protocol parameters



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



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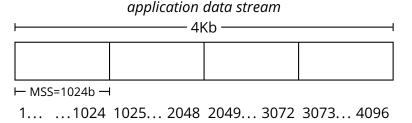
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4Kb	
— MSS=1024h →	

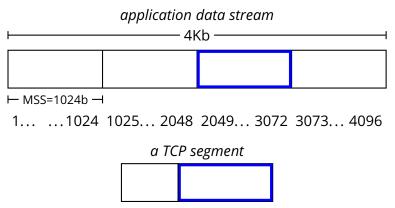
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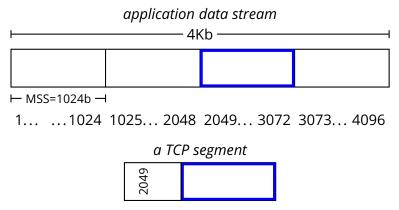
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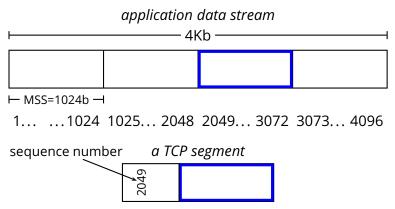
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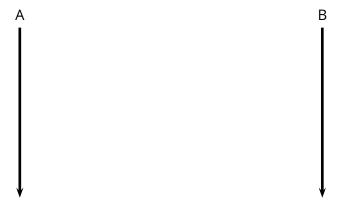
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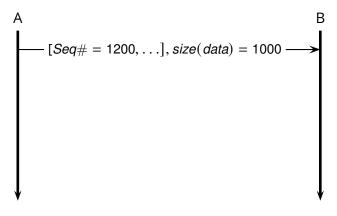


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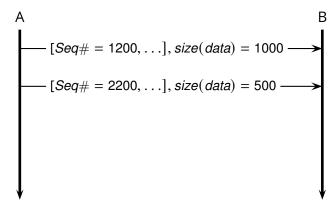
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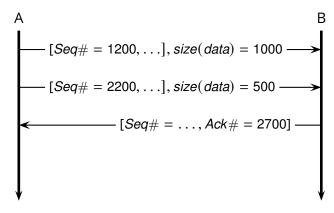
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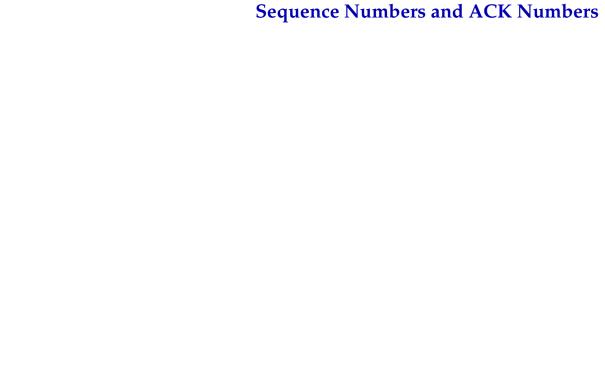


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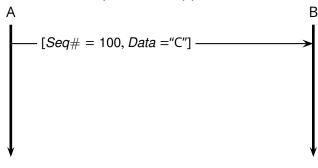


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 - two different sequence numbers

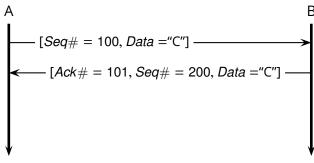
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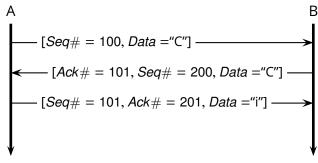
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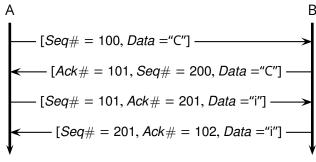
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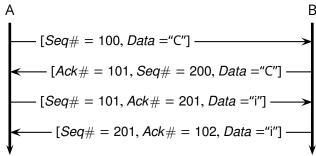


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E.g., consider a simple "Echo" application:



Acknowledgments are "piggybacked" on data segments

- Duplicate acknowledgments to detect lost segments
 - lacktriangleright receiver notices a missing packet ightarrow duplicate ACKs ightarrow retransmission by sender
- A *timer* to detect lost segments
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- 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)

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- 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

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r_send(data)
if (timer not running)
  start_timer()
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u_recv([ACK,y])

if (y > base)

base ← y

if (there are pending segments)

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else ...
```



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 - Immediate ACK: immediately send ACK if the packet start at the lower end of the gap



Reaction to ACKs (Sender)

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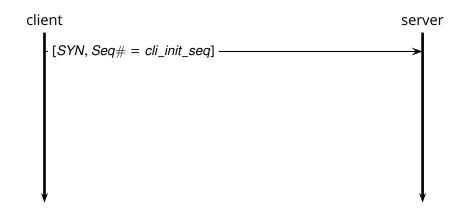
```
if (y > base)
base ← y
if (there are pending segments)
start_timer()
```

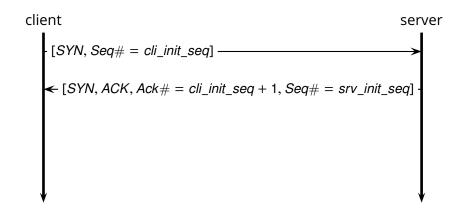
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  ack_counter[y] ← ack_counter[y] + 1
  if (ack_counter[y] = 3)
    u_send(segment with sequence number y)
```

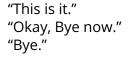






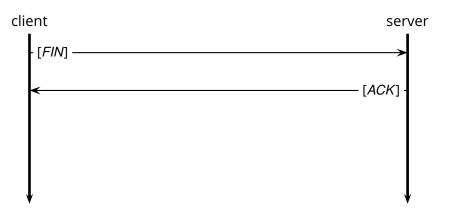


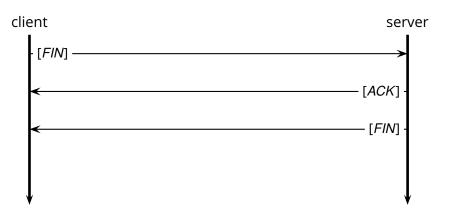
```
Se - [SYN, Seq\# = cli\_init\_seq] \longrightarrow
\leftarrow [SYN, ACK, Ack\# = cli\_init\_seq + 1, Seq\# = srv\_init\_seq] -
- [ACK, Seq\# = cli\_init\_seq + 1, Ack\# = srv\_init\_seq + 1] \longrightarrow
client
                                                                                                                                                                        server
```

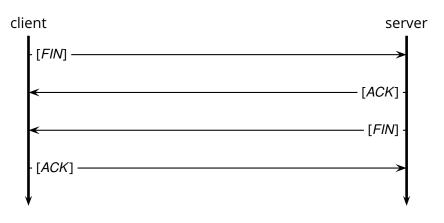




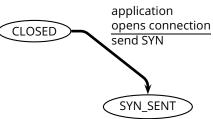


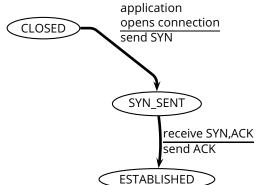




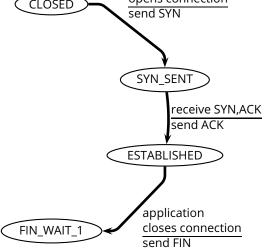


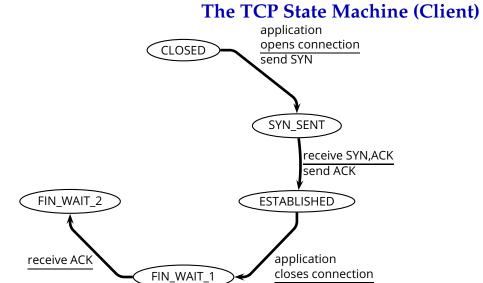






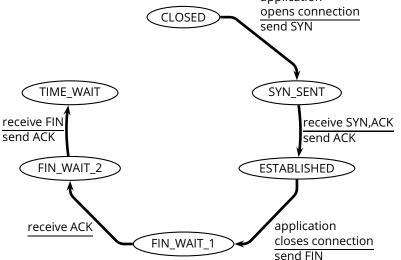
The TCP State Machine (Client) application opens connection send SYN

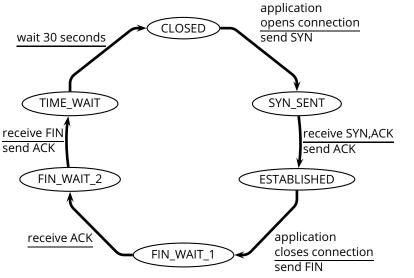




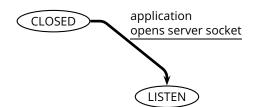
send FIN

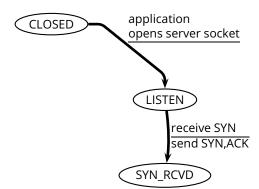
The TCP State Machine (Client) application opens connection CLOSED send SYN

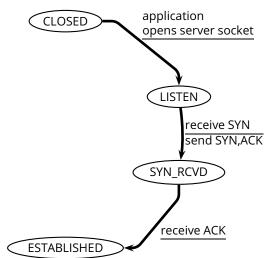


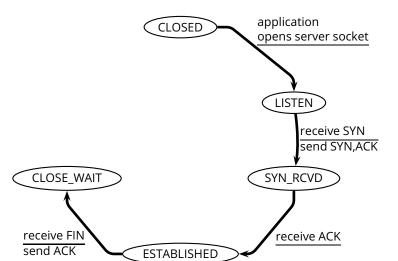


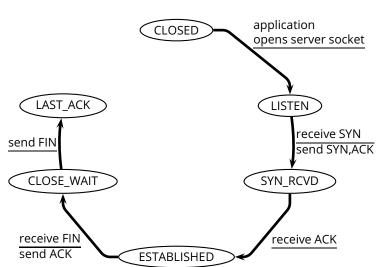


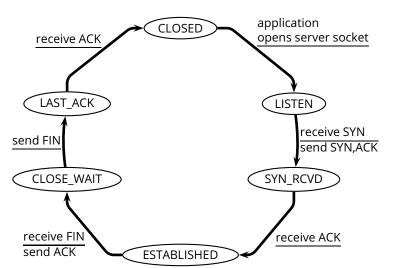




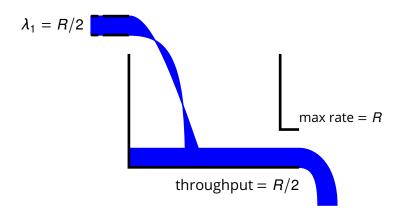


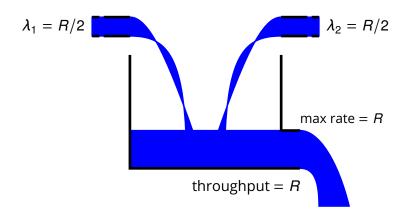


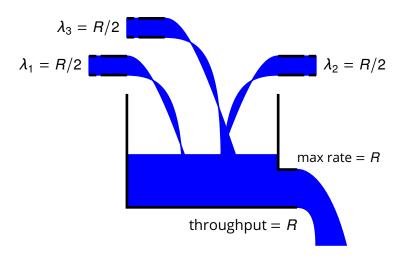


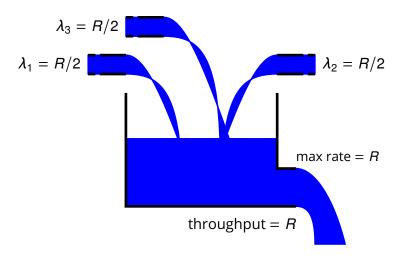


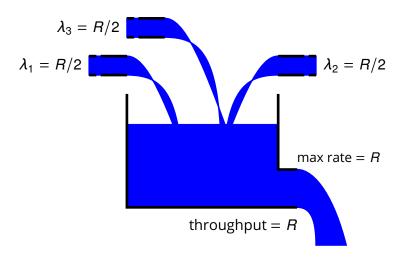


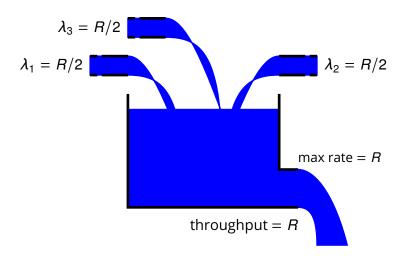


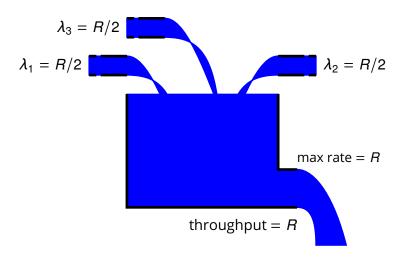














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(ideal input flow)

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

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

■ Steady-state queuing delay

ideal input flow λ_{in} constant

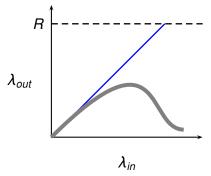
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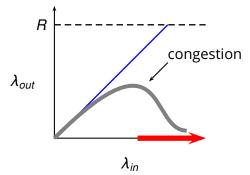
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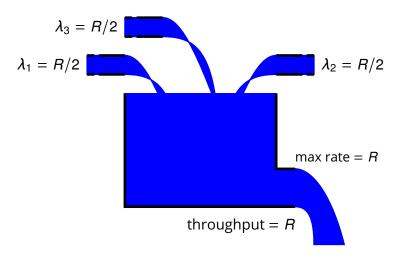
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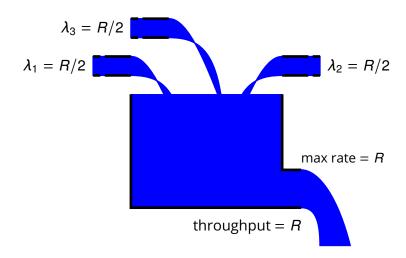


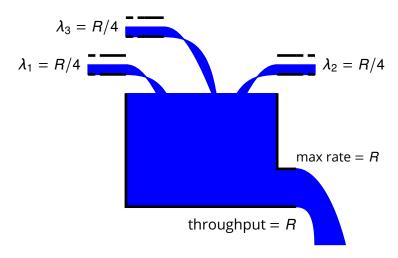
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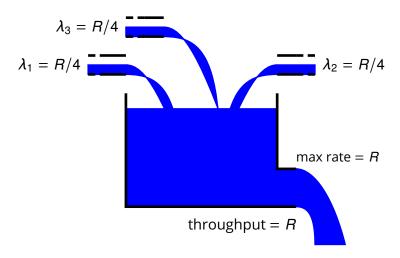


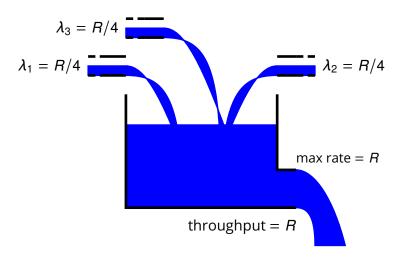
■ What to do when the network is congested?

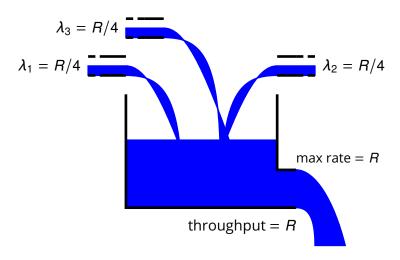












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- 3. how should the sender *control its output rate*?
 - we need a brain and we need to know how to drive!



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Detecting Congestion (Eyes)

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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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LastByteSent − LastByteAcked ≤ W

where

 $W = \min(CongestionWindow, ReceiverWindow)$

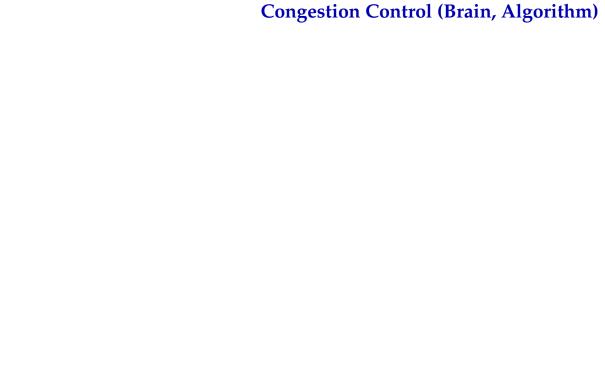
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■ The resulting maximum output rate is roughly

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



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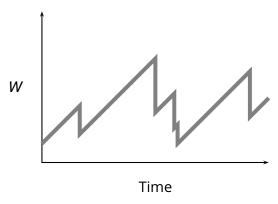
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 - e.g., suppose W = 14600 and MSS = 1460, then the sender increases W to 16060 after 10 acknowledgments acknowledgments

■ 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

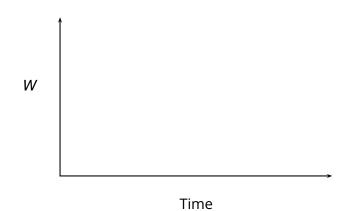
Assuming the current window size is $W = \overline{W}$

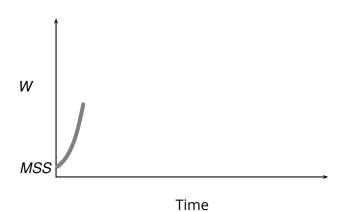
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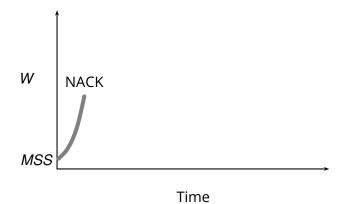
- Timeout
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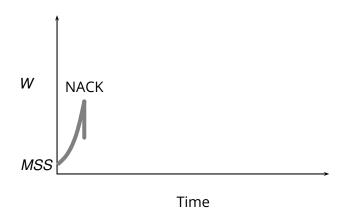
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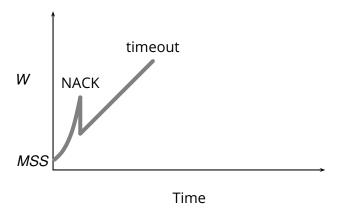
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- *NACK* (i.e., triple duplicate-ack)
 - ightharpoonup set $ssthresh = \overline{W}/2$
 - ightharpoonup cut W in half: $W = \overline{W}/2$
 - run *congestion avoidance*, ramping up *W* linearly
 - ► This is called *fast recovery*

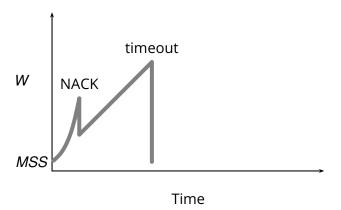


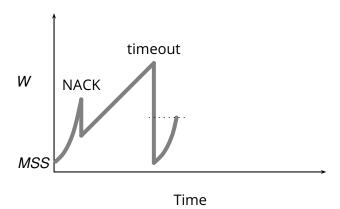


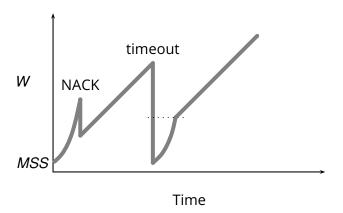


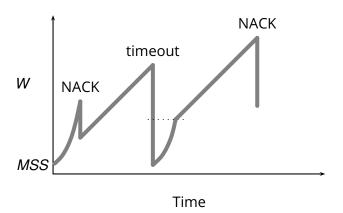


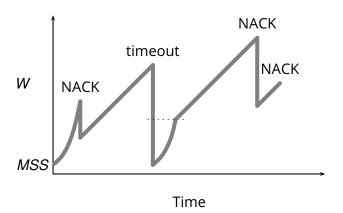


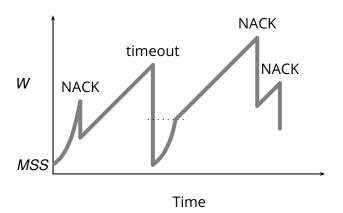


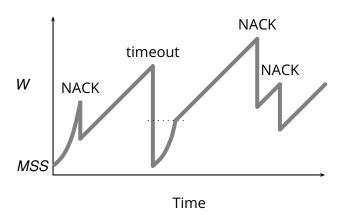


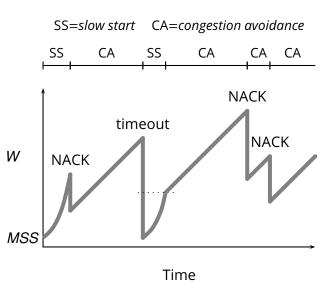




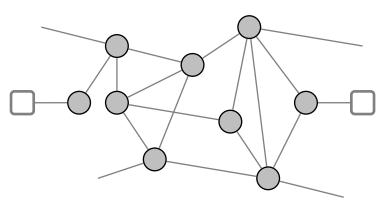


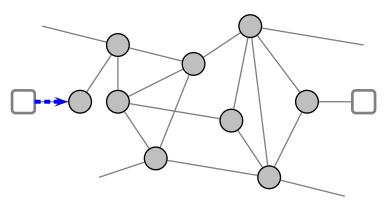


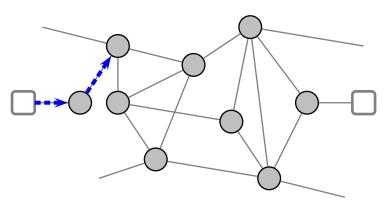


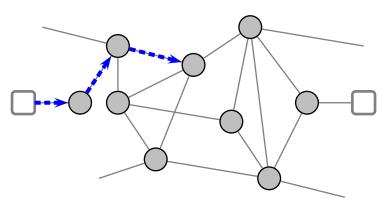


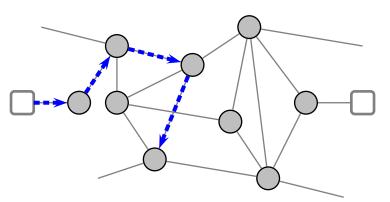
Datagram Network

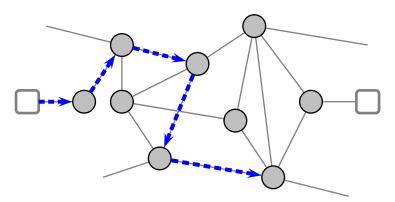


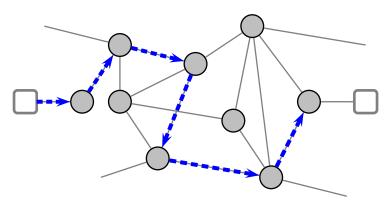


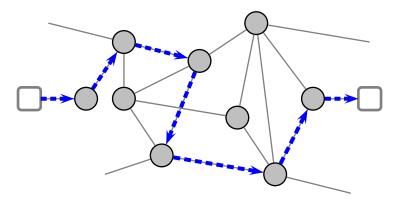


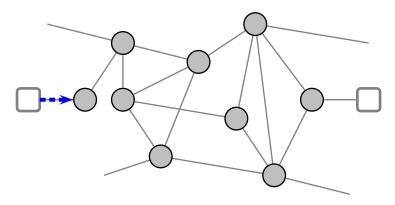


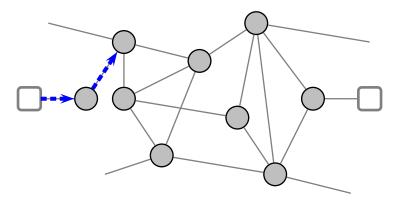


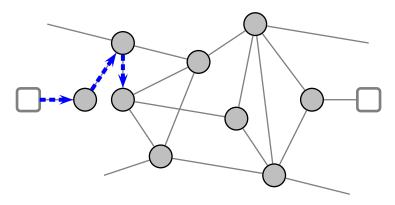


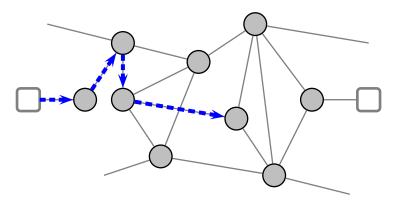


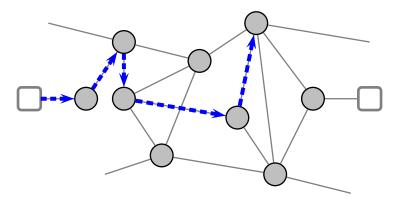


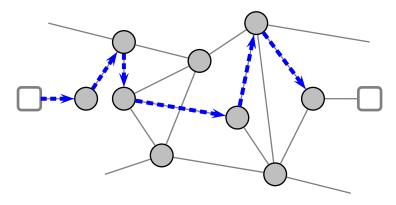


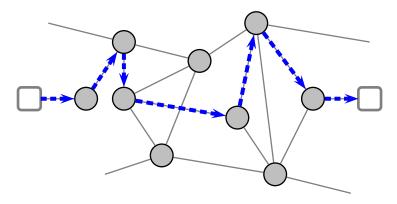


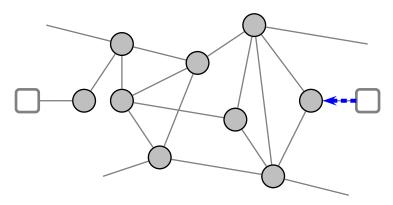




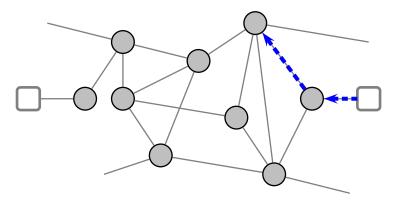




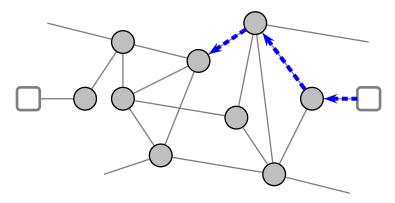




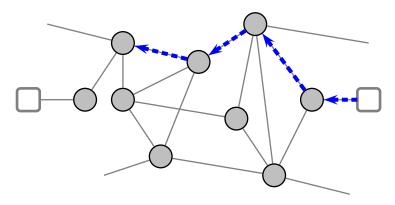
- Potentially *multiple paths* for the same source/destination
- Potentially *asymmetric paths*



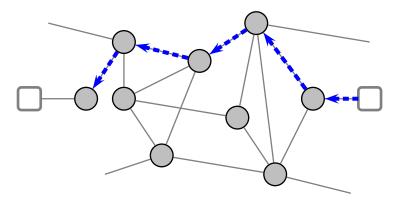
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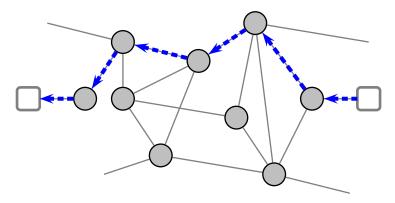
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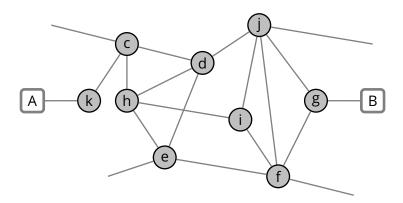


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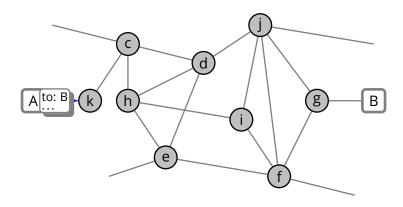


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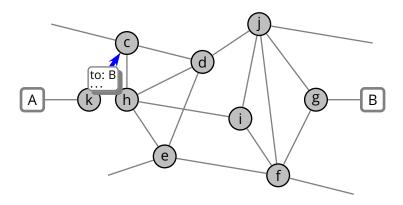




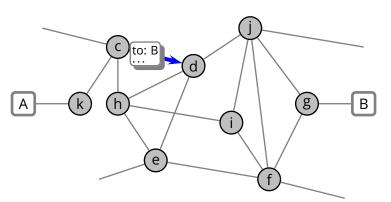
■ *A* sends a datagram to *B*

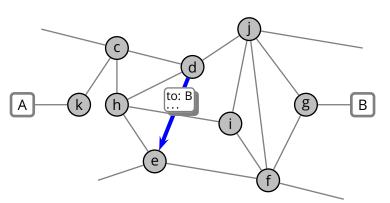


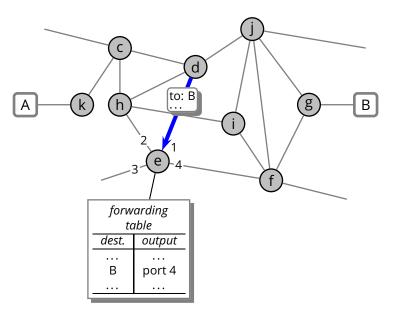
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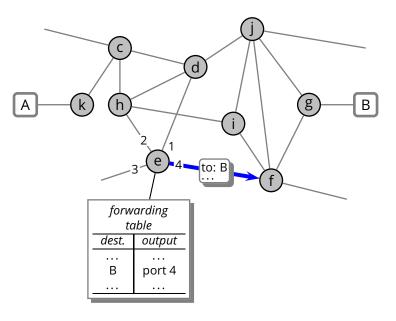


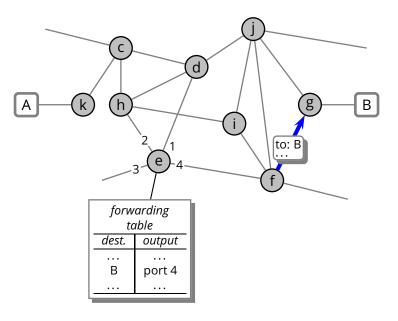
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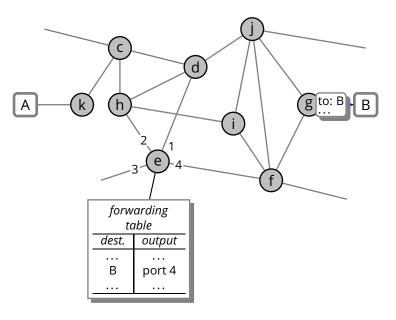












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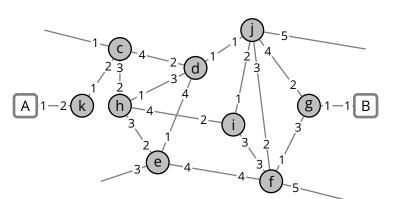
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 - how fast does the router have to forward datagrams?

Forwarding

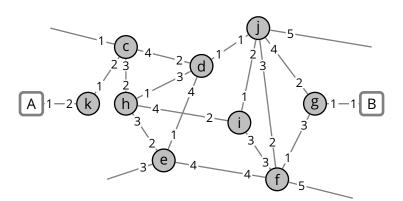
- Input: datagram destination
- Output: output port
- Simple design: "forwarding table"
- Issues
 - how big is the forwarding table?
 - how fast does the router have to forward datagrams?
 - how does the router build and maintain the forwarding table?



Routing



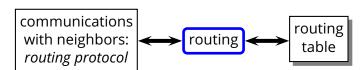
Routing



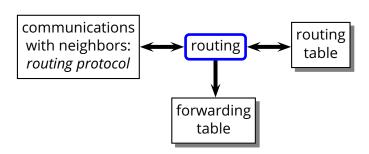
router k		
Α	2	ļ
В	1	



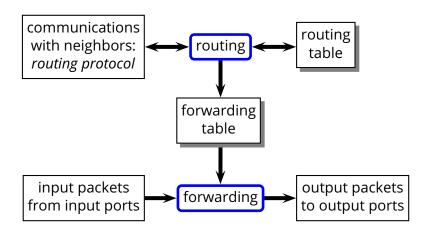
Router Functions



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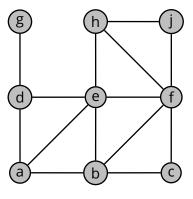


Routing Problem

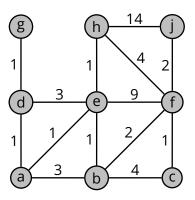
■ Finding paths through a network

Routing Problem

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■ Finding paths through a network



■ Example: $a \rightarrow j$?

Graph Model

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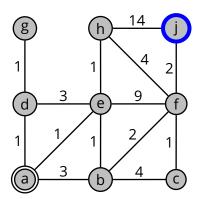
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- Compile *u*'s forwarding table by adding the following entry:

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

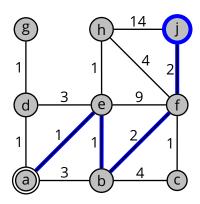
- \blacktriangleright A(v) is the address (or set of addresses) of router v
- ► $I_u(x_1)$ is the interface that connects u to the first next-hop router x_1 in $P_{u \to v} = u, x_1, x_2, \dots, x_n, v$

Back To The Example



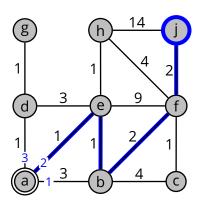
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 - least-cost path is $P_{a \rightarrow j} = a, e, b, f, j$
 - a's forwarding table will contain an entry $j \rightarrow 2$ since $l_a(e) = 2$

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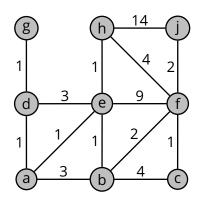
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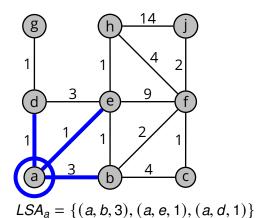
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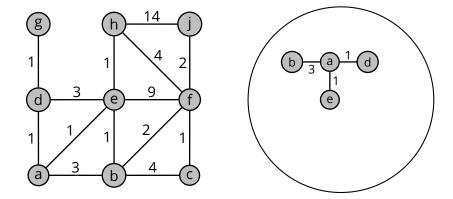
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Link-State Advertisements

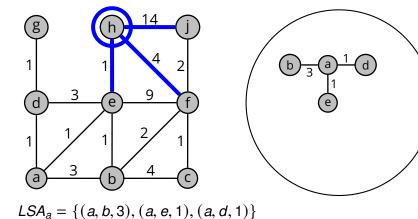


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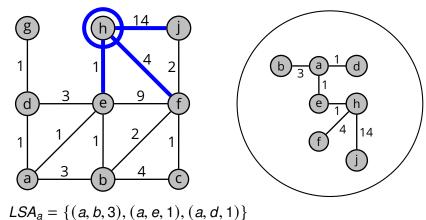




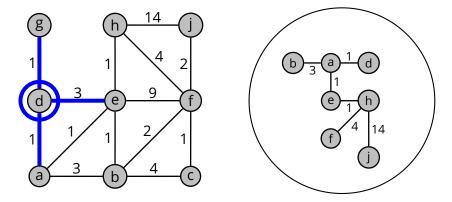
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 $LSA_h = \{(h, e, 1), (h, f, 4), (h, j, 14)\}$



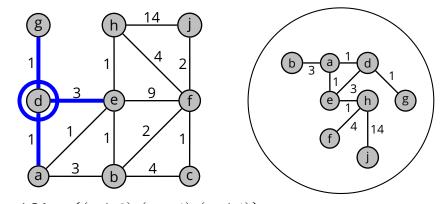
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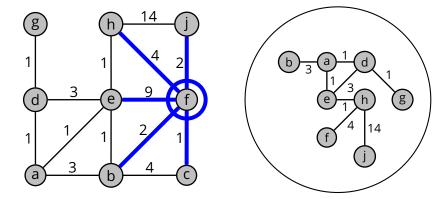
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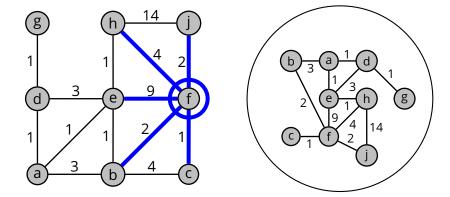


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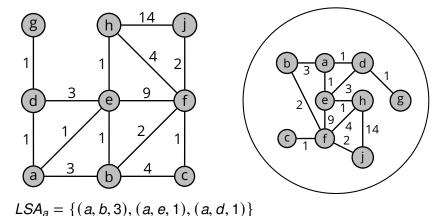


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- Every router sends its LSA to every other router in the network, so we need a **broadcast routing scheme**
- Once we have all the LSAs from every router, and therefore we complete knowledge of *G*, we need an *algorithm to compute least-cost paths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



■ Flooding

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 - it requires (unicast) routing information
 - so it is obviously useless to implement a routing algorithm

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



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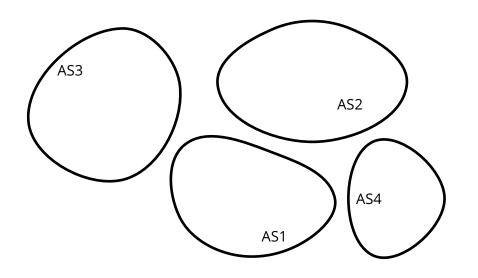
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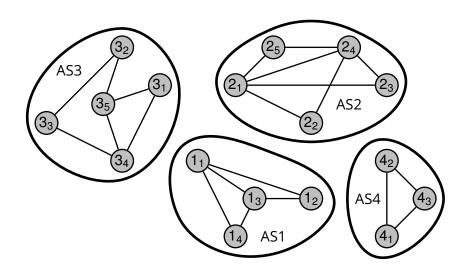
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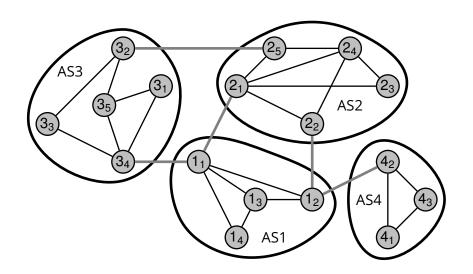
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- An intra-autonomous system routing protocol runs within an autonomous system (e.g., OSPF)
 - this protocol determines internal routes
 - internal router ↔ internal router
 - internal router ↔ gateway router
 - ▶ gateway router ↔ gateway router

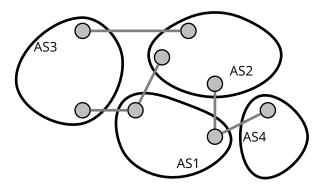






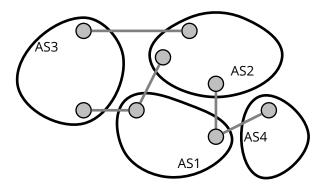
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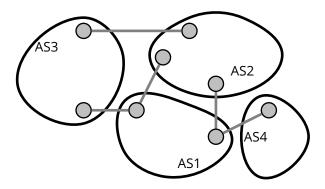
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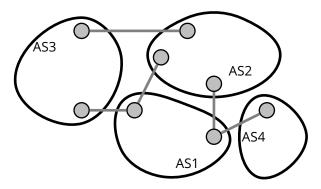
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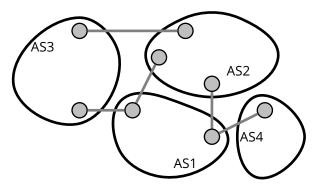
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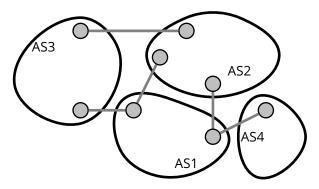
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- Both inter-AS and intra-AS routing information is used to compile the forwarding tables

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 - what if x is reachable through multiple gateway routers G_x , G_x' , . . .?
 - 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
 - this process is called supernetting



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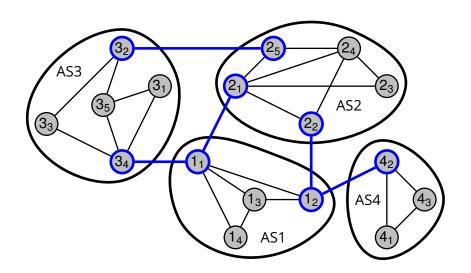
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- BGP internal session (iBGP): a session within an autonomous system
 - note that internal sessions carry inter-AS information
 - intra-AS routing uses a separate protocol (e.g., OSPF)

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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