

# Lecture 16: Network Layer Overview, Internet Protocol

COMP 332, Fall 2018

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**Acknowledgements:** materials adapted from Computer Networking: A Top Down Approach 7<sup>th</sup> edition: ©1996-2016, J.F Kurose and K.W. Ross, All Rights Reserved as well as from slides by Abraham Matta at Boston University, and some material from Computer Networks by Tannenbaum and Wetherall.

# Today

## 1. Announcements

- homework 6 posted
  - discuss: UDP ping server, chat server + reliability
- midterm graded, will hand back once everyone has written it

## 2. TCP congestion control

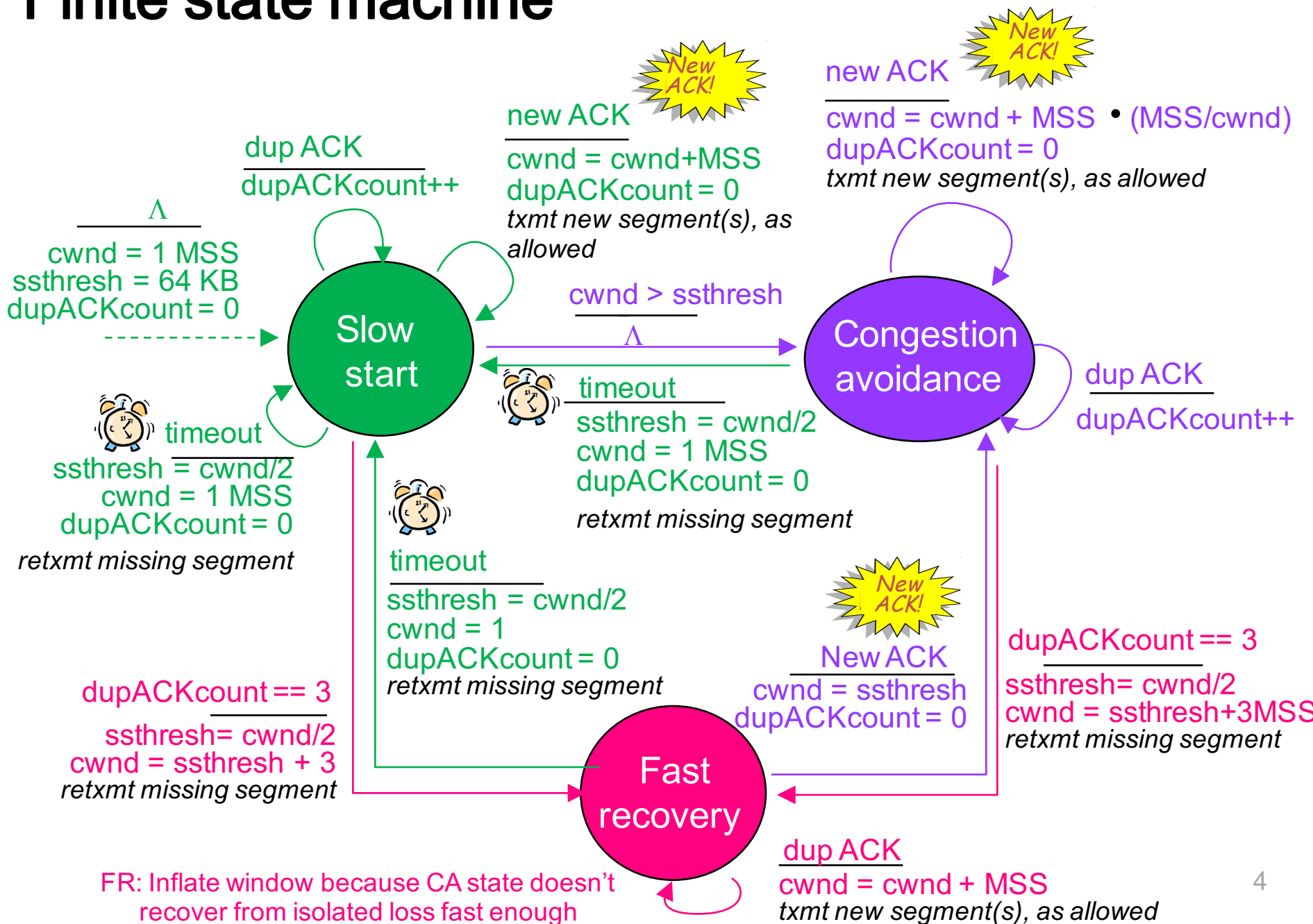
## 3. Network layer

- overview
- what's inside a router
- Internet protocol (IP)

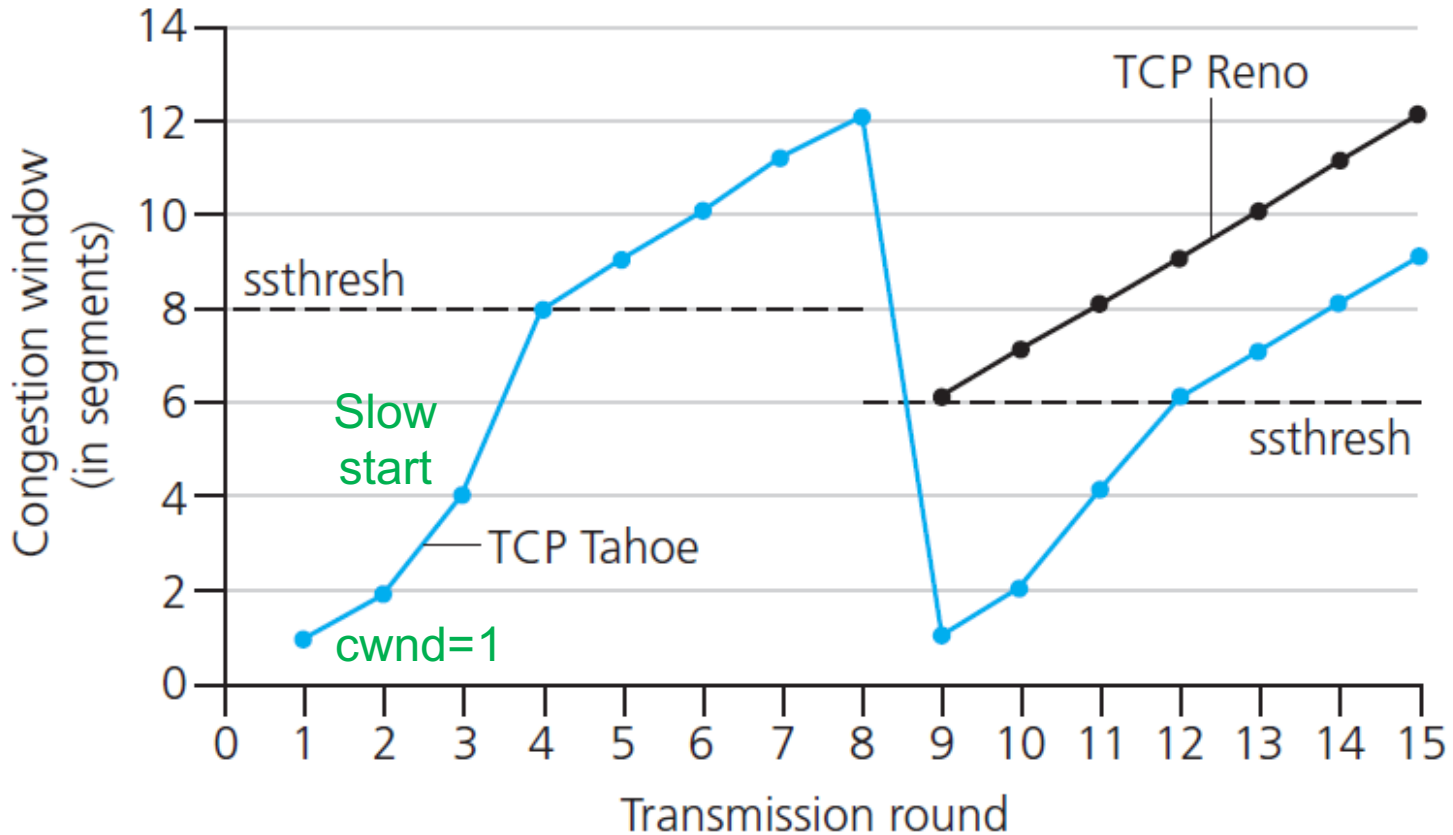
**TCP**

# **CONGESTION CONTROL**

# Finite state machine



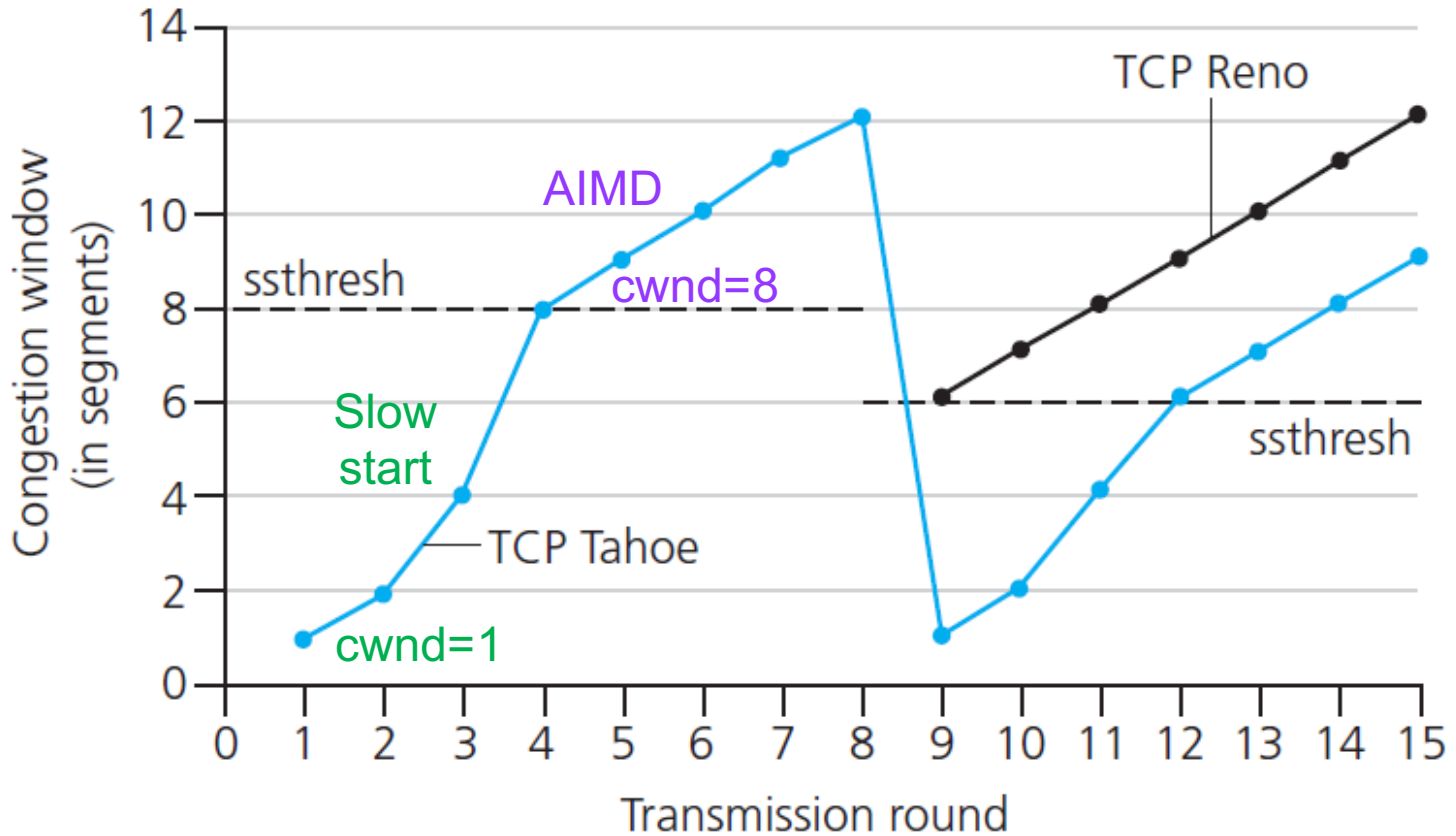
# Slow start: when to stop exponential increase?



## Slow start

- initially  $cwnd = 1$  MSS
- every time ACK received, double  $cwnd$

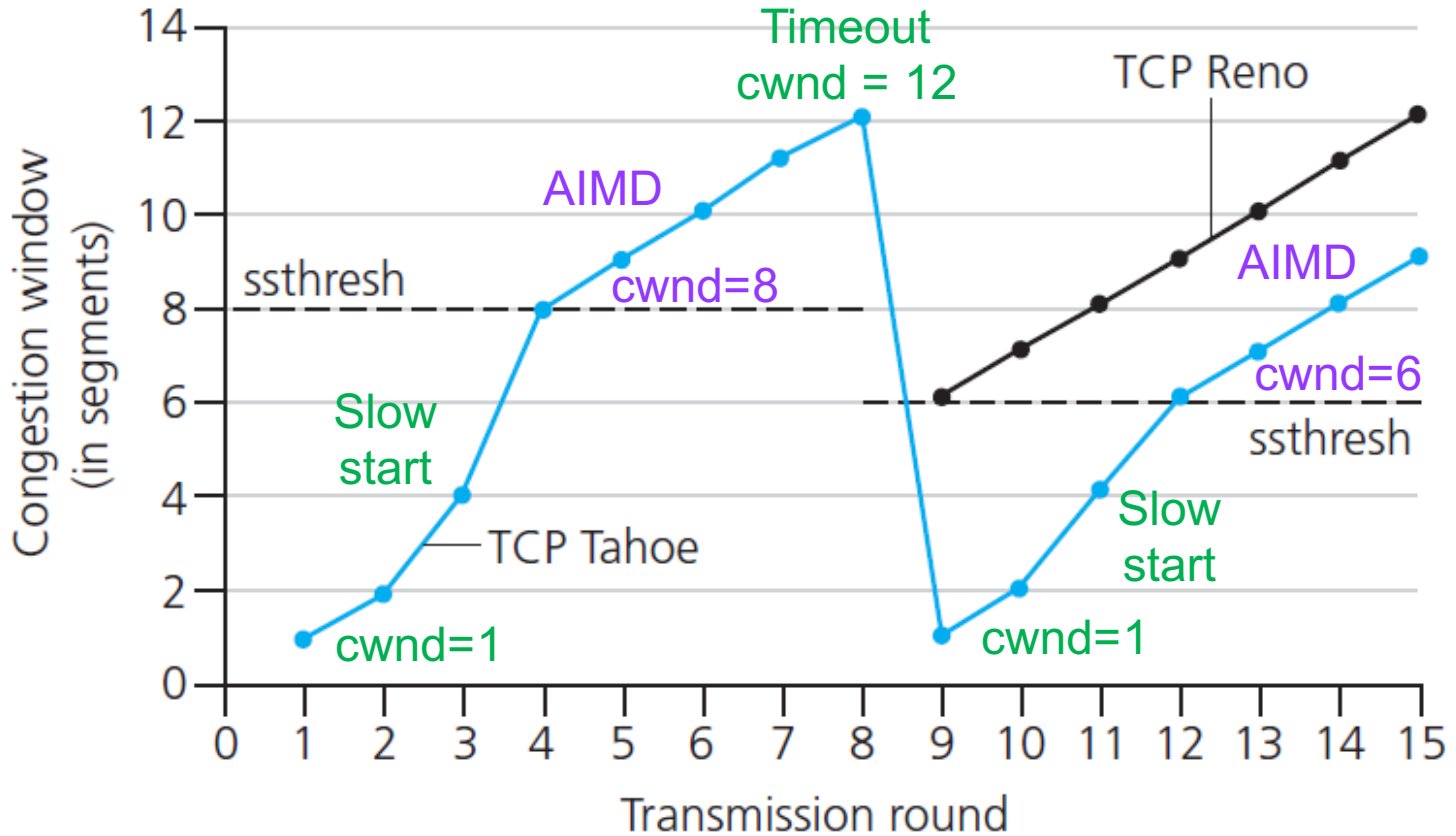
# Congestion avoidance



When  $cwnd = ssthresh$

- go to congestion avoidance
- use AIMD

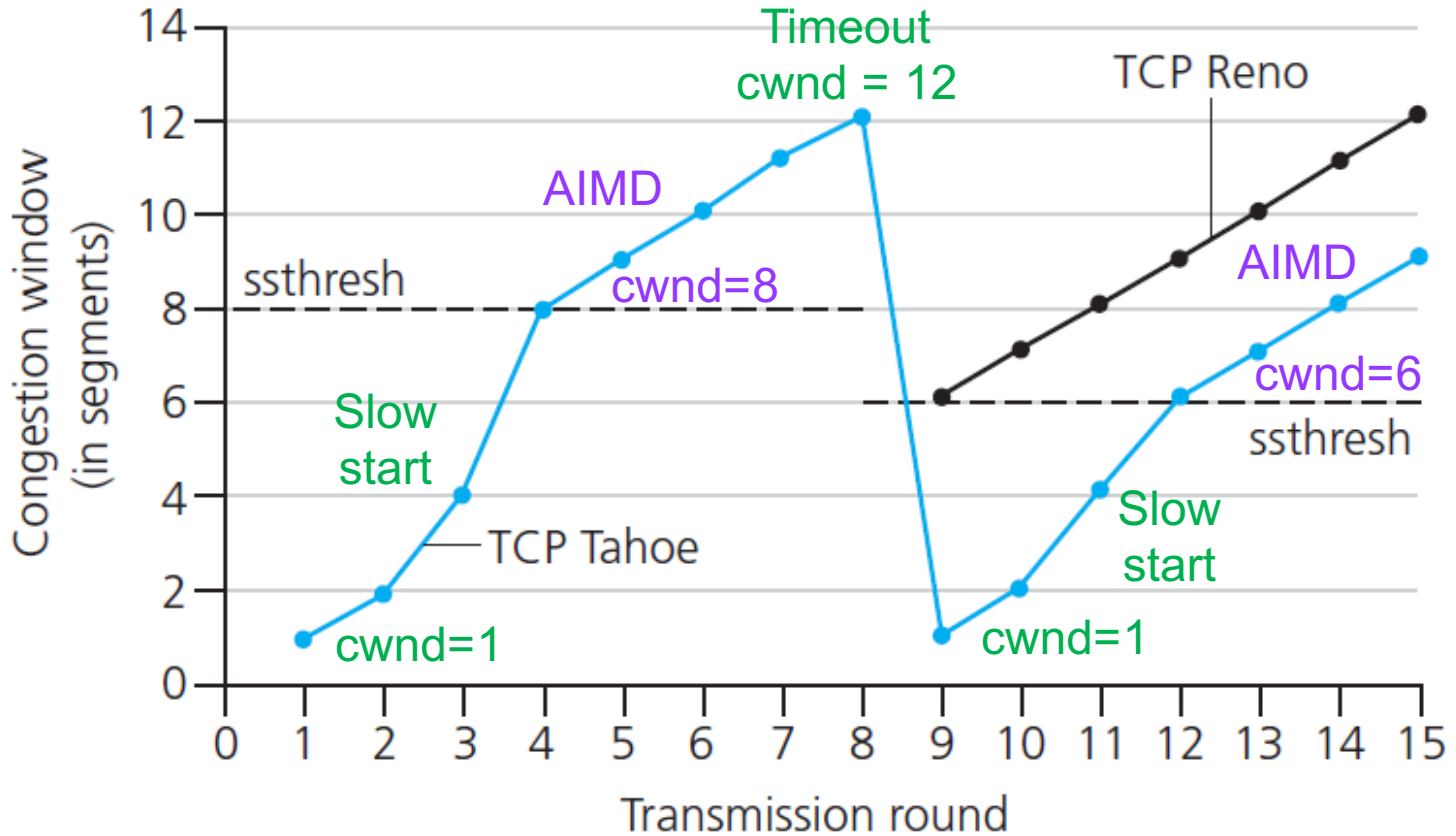
# Timeout



## Restart slow start when timeout

- $ssthresh = cwnd/2$
- $cwnd = 1$  MSS

# 3 duplicate ACKs



If 3 duplicate ACKs go to fast recovery

- $ssthresh = cwnd/2$
- $cwnd = ssthresh + 3 \text{ MSS}$



# Average TCP throughput

## Focus on AIMD

- ignore slow start, assume always data to send

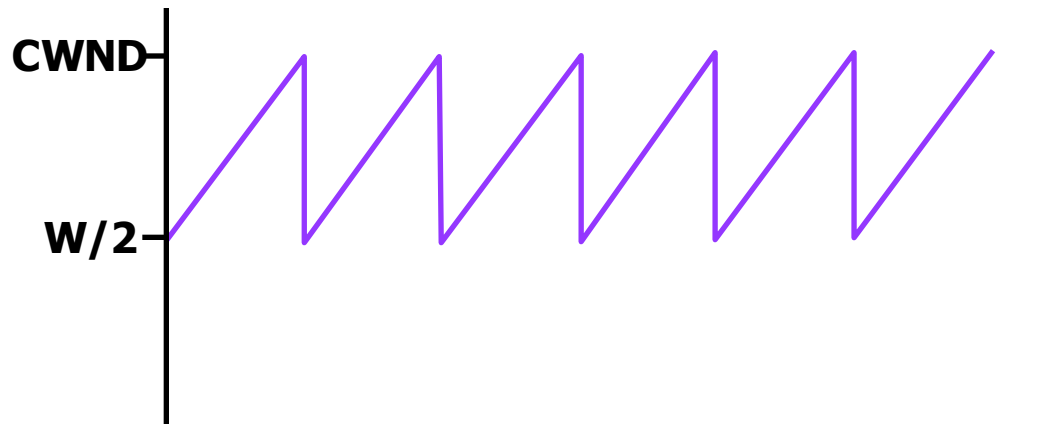
### Max rate

- $\text{cwnd} / \text{RTT}$

### 3 dup loss rate

- $0.5 \text{ cwnd} / \text{RTT}$

$$\text{Avg TCP thrupt} = \frac{3}{4} \frac{\text{CWND}}{\text{RTT}} \text{ bytes/sec}$$



# Setting window size

Window is min (rwnd, cwnd)

```
▼ Transmission Control Protocol, Src Port: 443 (443), Dst Port: 52232 (52232), Seq: 0, Ack: 1,
  Source Port: 443
  Destination Port: 52232
  [Stream index: 0]
  [TCP Segment Len: 0]
  Sequence number: 0 (relative sequence number)
  Acknowledgment number: 1 (relative ack number)
  Header Length: 32 bytes
  ▼ Flags: 0x012 (SYN, ACK)
    000. .... = Reserved: Not set
    ...0 .... = Nonce: Not set
    .... 0... = Congestion Window Reduced (CWR): Not set
    .... .0.. = ECN-Echo: Not set
    .... ..0. = Urgent: Not set
    .... ...1 .... = Acknowledgment: Set
    .... .... 0... = Push: Not set
    .... .... .0.. = Reset: Not set
    ► .... .... ..1. = Syn: Set
    .... .... ...0 = Fin: Not set
    [TCP Flags: *****A**S*]
    Window size value: 8190
    [Calculated window size: 8190]
```

rwnd

# Network Layer

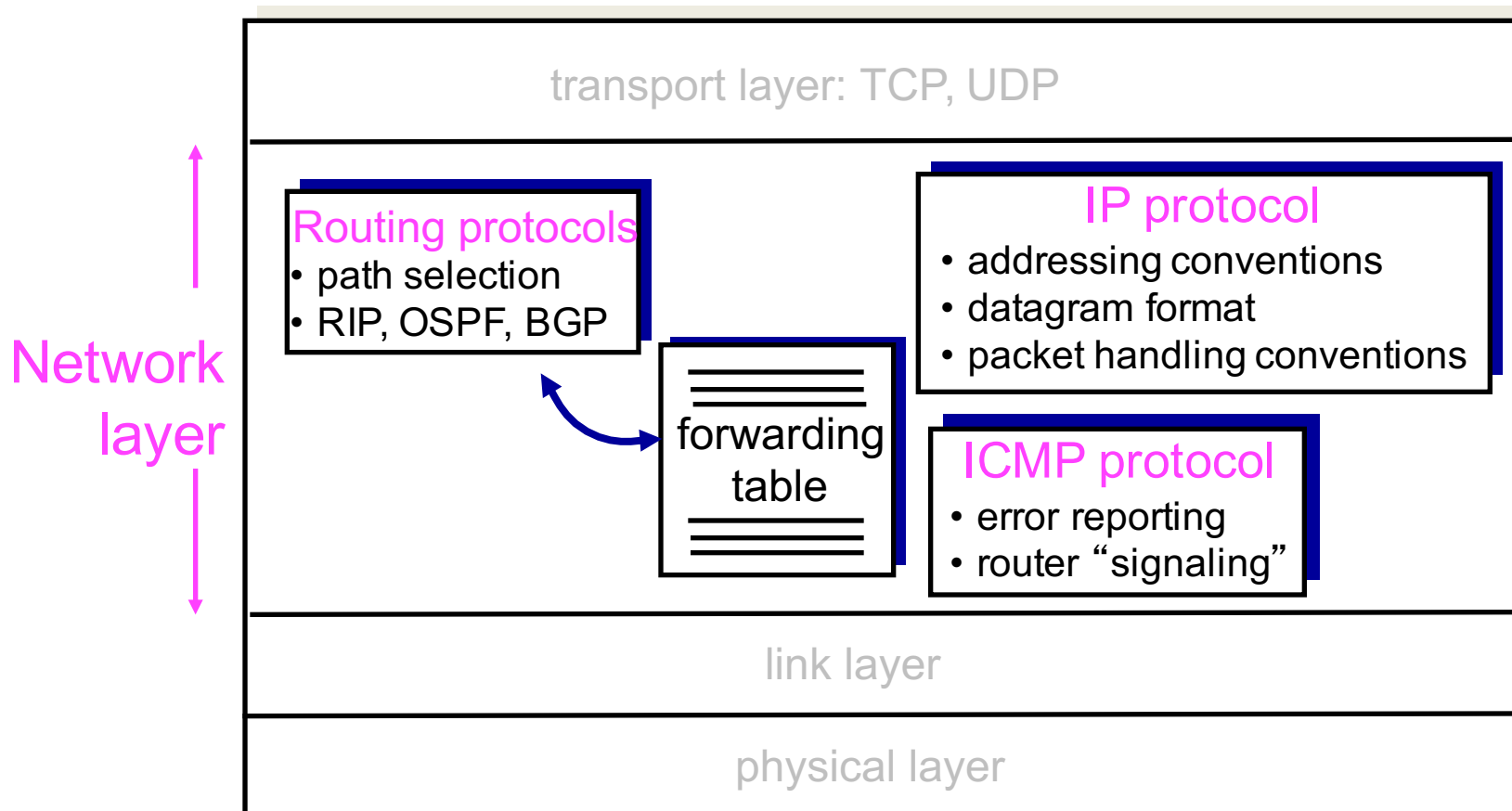
## **OVERVIEW**

# 5-layer Internet protocol stack

Layer	Service provided to upper layer	Protocols	Unit of information
5 Application	<ul style="list-style-type: none"><li>Support network applications</li></ul>	FTP, DNS, SMTP, HTTP	<b>Message</b> 1 message may be split into multiple segments
4 <b>Transport</b>	<ul style="list-style-type: none"><li>Deliver messages to app endpoints</li><li>Flow control</li><li>Reliability</li></ul>	TCP (reliable) UDP (best-effort)	<b>Segment</b> (TCP) <b>Datagram</b> (UDP) 1 segment may be split into multiple packets
3 <b>Network</b>	<ul style="list-style-type: none"><li>Route segments from source to destination host</li></ul>	IP (best-effort) Routing protocols	<b>Packet</b> (TCP) <b>Datagram</b> (UDP)
2 <b>Link</b>	<ul style="list-style-type: none"><li>Move packet over link from one host to next host</li></ul>	Ethernet, 802.11	<b>Frame</b> MTU is 1500 bytes
1 <b>Physical</b>	<ul style="list-style-type: none"><li>Move individual bits in frame from one host to next</li><li>“bits on wire”</li></ul>	Ethernet phy 802.11 phy Bluetooth phy DSL	<b>Bit</b>

# Internet's network layer

## Network layer functions on hosts and routers



# Network layer

## Goal

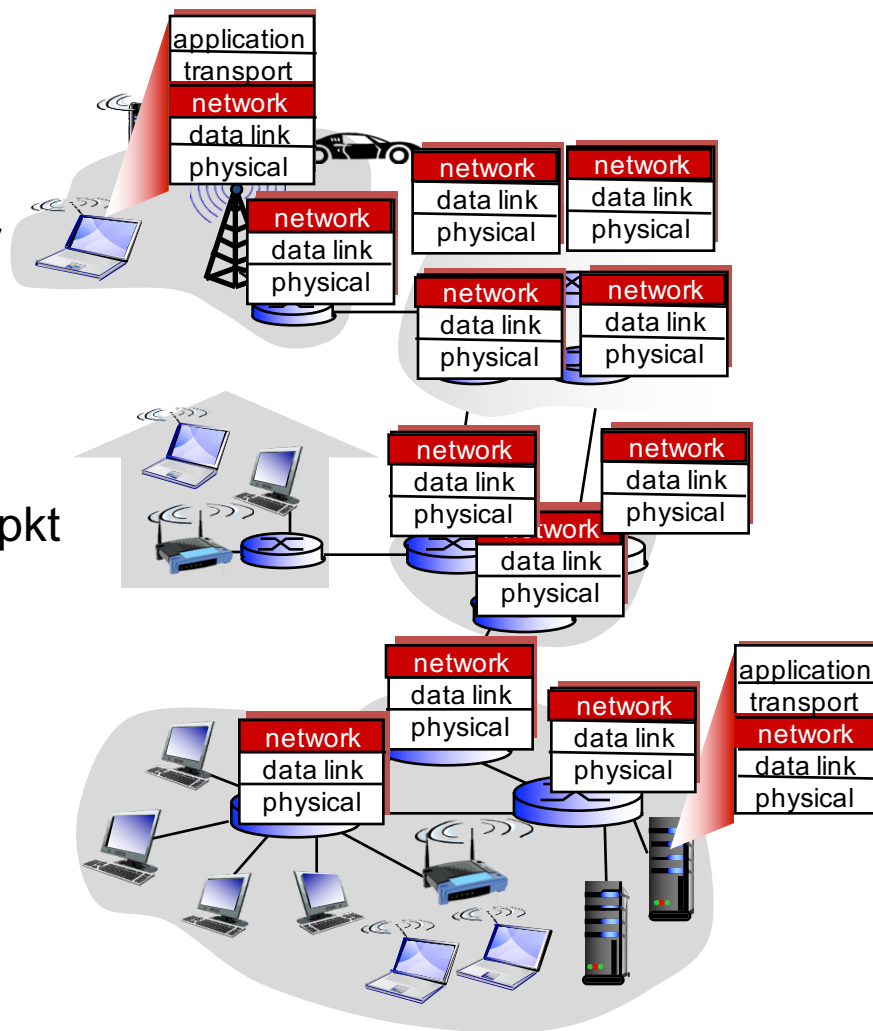
- move pkt from one host to another

## How done on Internet?

- routers
  - examine header fields in every IP pkt
  - determines outgoing link

## Internet e2e argument

- some functionality only properly implemented in end systems
- smart hosts vs. dumb routers



Network layer is in every host and router on Internet

# Encapsulation and decapsulation

## Sender

- encapsulates segments into packets, puts src, dest IP in IP pkt hdr

## Receiver

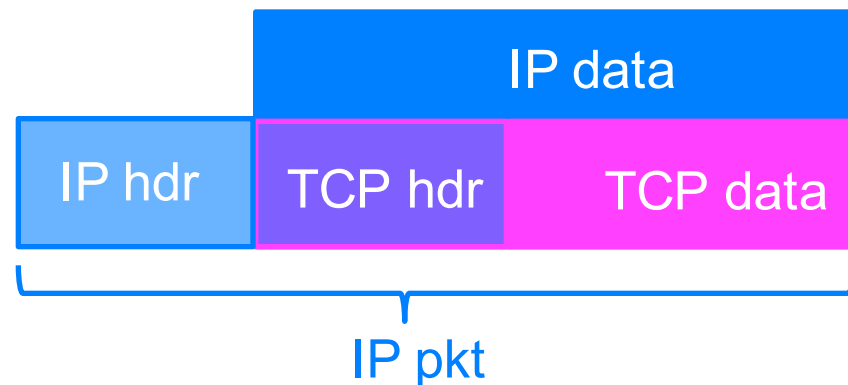
- decapsulates packets into segments, delivers to transport layer

## Max length of IP packet in bytes

- MTU: Maximum Transmission Unit
- 1500 bytes if Ethernet used as link layer protocol

## Max length of TCP data in bytes

- MSS: Maximum Segment Size
- $MSS = MTU - IP\ hdr - TCP\ hdr$ 
  - TCP header  $\geq 20$ bytes



# Division of network layer functionality

## 1. Control plane

- comprises traffic only between routers, to compute routes between src and dst
- network-wide: routers run routing algorithms

## 2. Data plane

- comprises traffic between end hosts, forwarded by routers
- forwarding table set based on routes computed in control plane
- local: each router stores, forwards packets



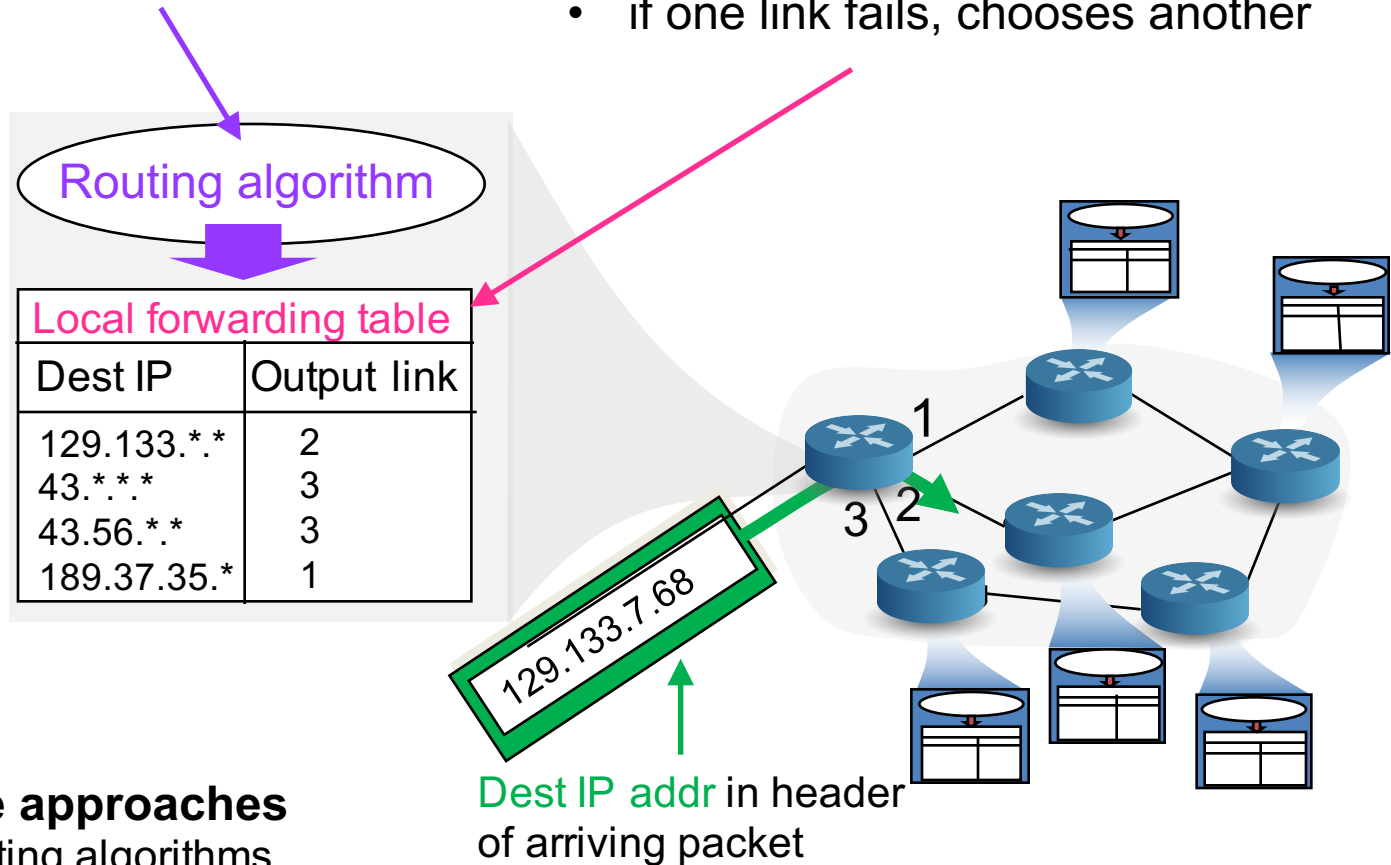
# Interplay between routing and forwarding

## Routing (slower time scale)

- routers view Internet as **graph**
- run **shortest path algorithms**

## Forwarding (faster time scale)

- routers use paths to choose best **output link** for packet **destination IP address**
- if one link fails, chooses another

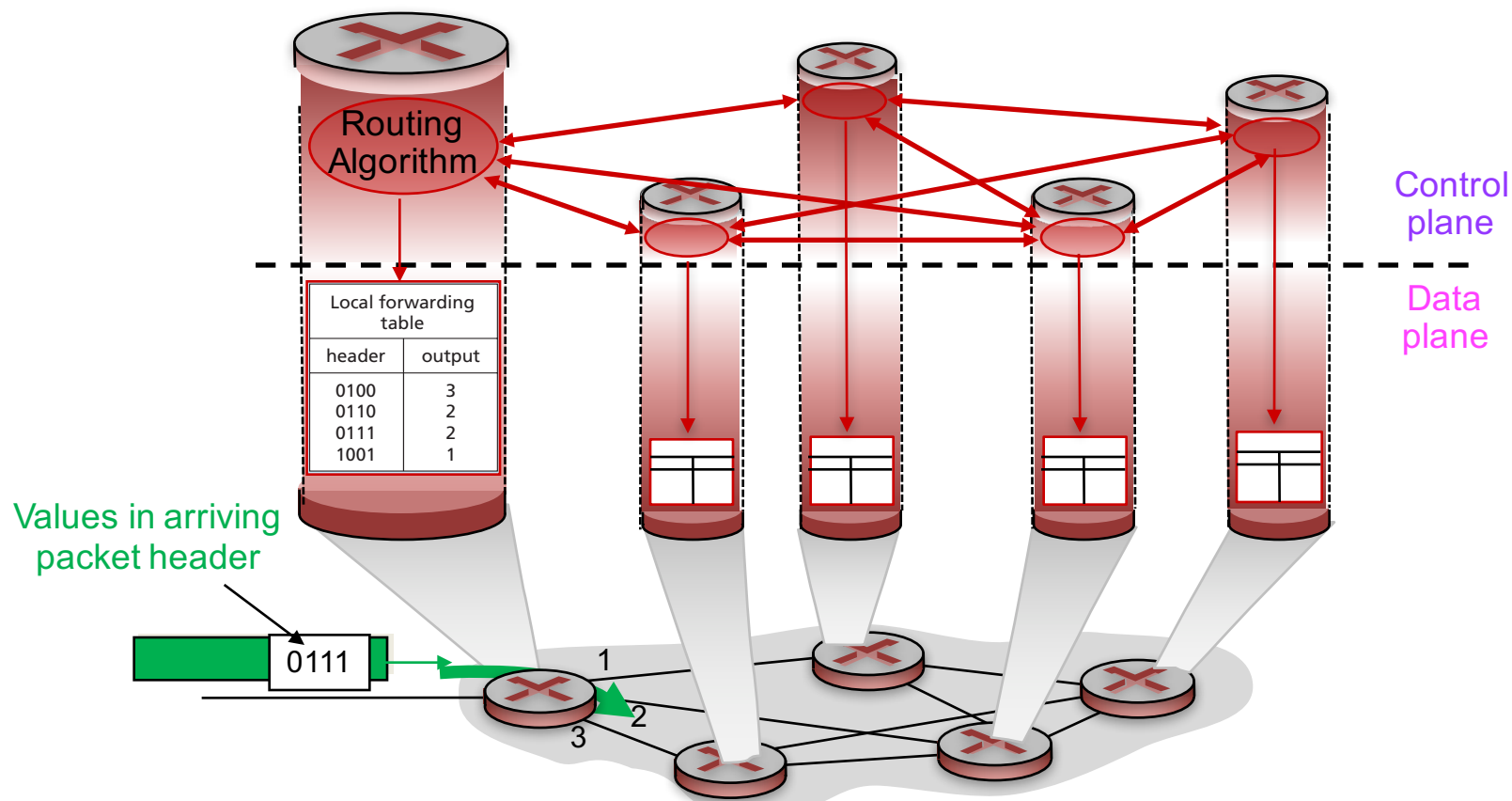


## 2 control-plane approaches

1. traditional routing algorithms implemented in routers
2. software-defined networking (SDN) implemented in (remote) servers

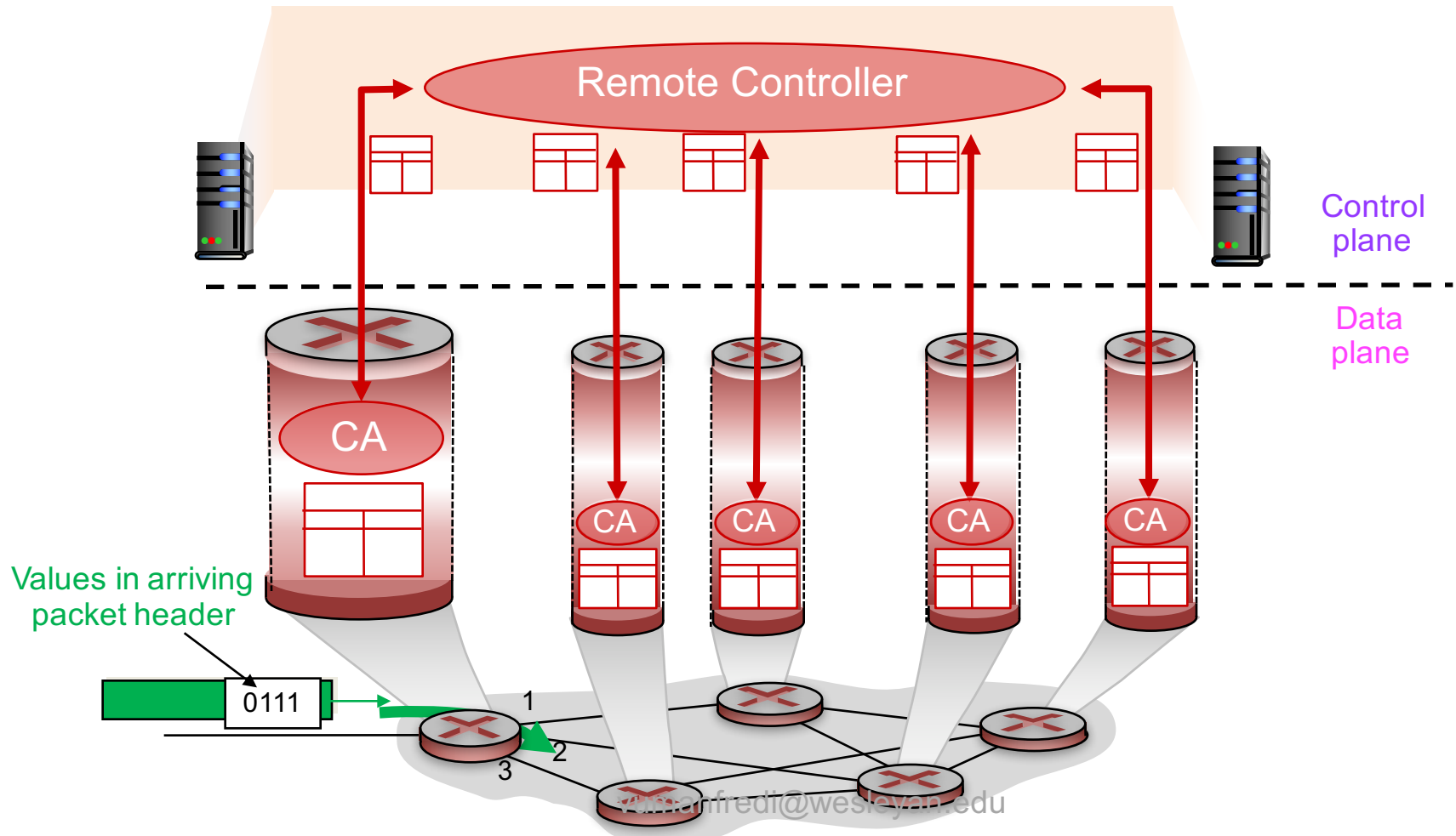
# Approach 1: per-router control plane

Individual routing algorithm components in each and every router interact in the control plane



# Approach 2: logically centralized control plane

A distinct (typically remote) controller interacts with local control agents (CAs)



# Network layer service model

Q: What **service model** does network layer provide to transport layer for moving packets from sender to receiver?

## Example services

- individual packets
  - guaranteed delivery
  - guaranteed delivery with less than 40 ms delay
- flow of packets
  - in-order packet delivery
  - guaranteed minimum bandwidth to flow
  - restrictions on changes in inter-packet spacing

# Network layer service models

Network Architecture	Service Model	Guarantees ?				Congestion feedback
		Bandwidth	Loss	Order	Timing	
Internet	best effort	none	no	no	no	no (inferred via loss)
ATM	CBR	constant rate	yes	yes	yes	no congestion
ATM	VBR	guaranteed rate	yes	yes	yes	no congestion
ATM	ABR	guaranteed minimum	no	yes	no	yes
ATM	UBR	none	no	yes	no	no

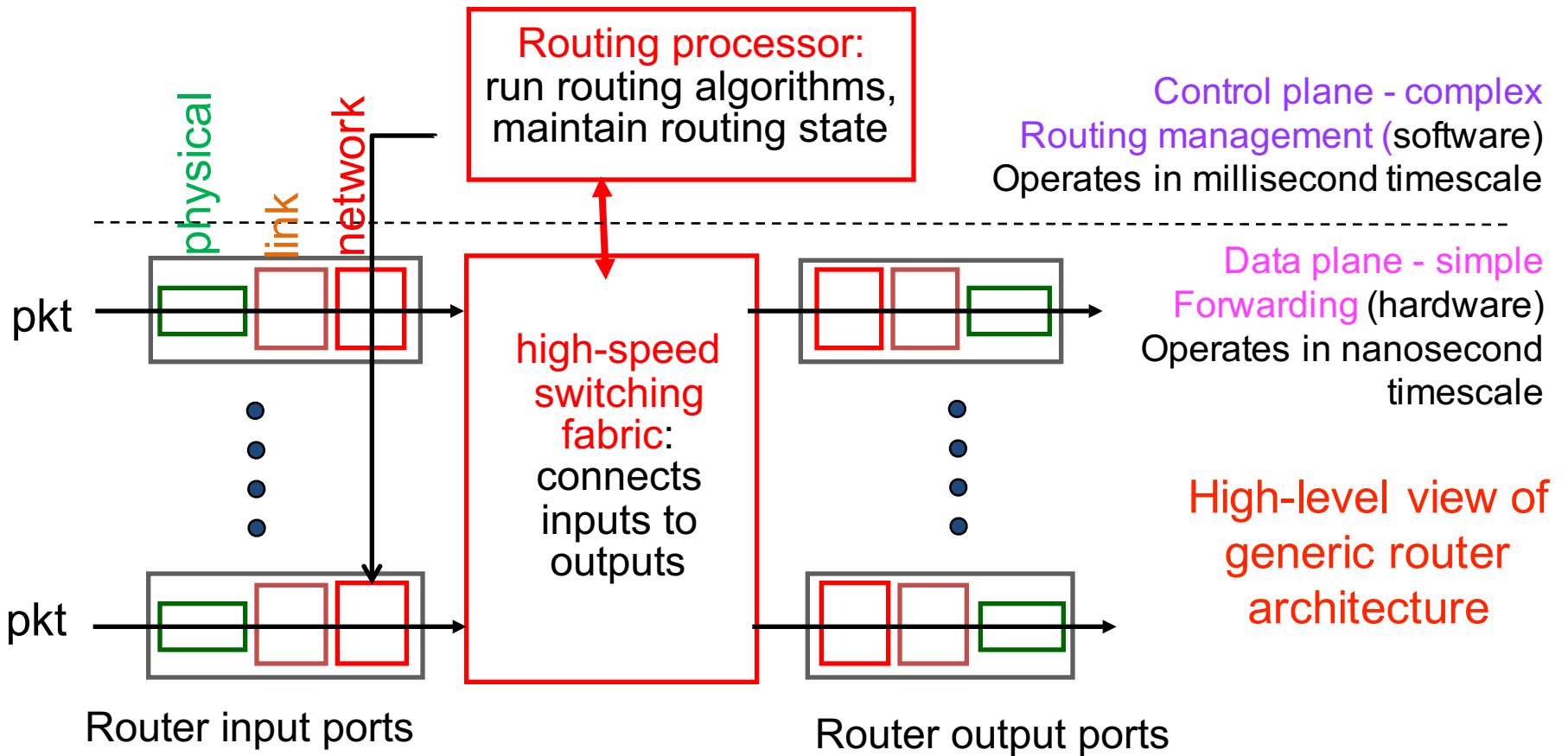
ATM: Asynchronous Transfer Mode  
 e.g., used in public switched telephone network

## Network Layer

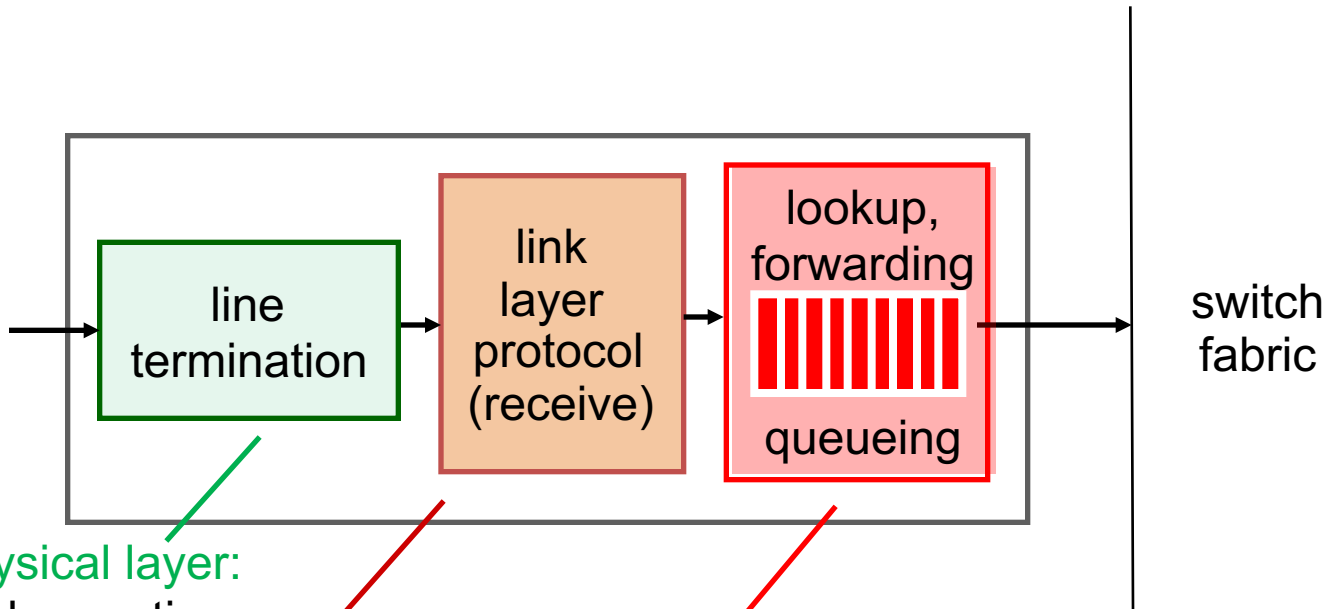
# WHAT'S INSIDE A ROUTER?

# What does a router need to do?

Run routing protocols (control) and store and forward pkts (data)



# Input port functions



Physical layer:  
bit-level reception,  
terminate phys. conn.

Data link layer:  
e.g., Ethernet processing,  
error-checking, de-capsulation,

Network layer

- validate/update checksum, decrement TTL
- **switching**: use header field values, lookup output port
- **queue**: if packets arrive faster than forwarding rate into switch fabric



# Switching fabrics

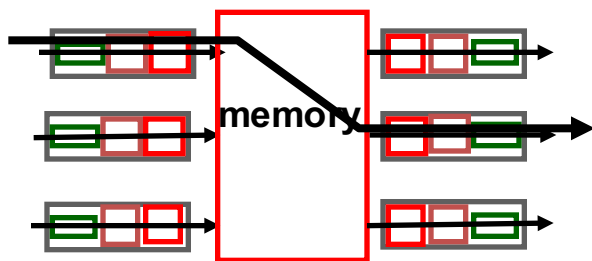
## Transfer packet

- from **input** buffer to appropriate **output** buffer

## Switching rate

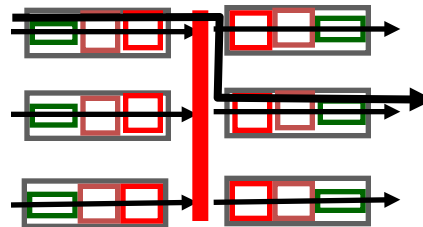
- rate at which packets can be **transferred** from inputs to outputs
- N inputs: switching rate = N x line rate desirable

## 3 types of switching fabrics



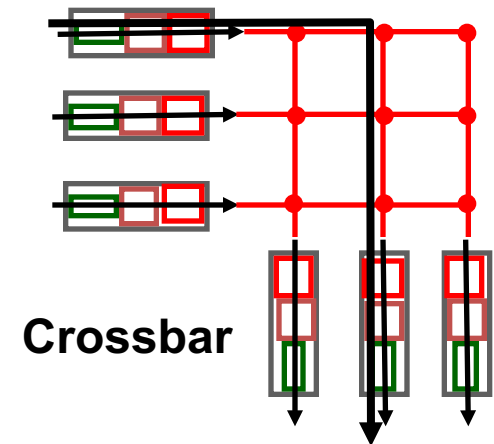
**Memory**

Speed limited by  
memory bandwidth



**Bus**

Speed limited by  
bus contention



**Crossbar**

Forward multiple  
pkts in parallel

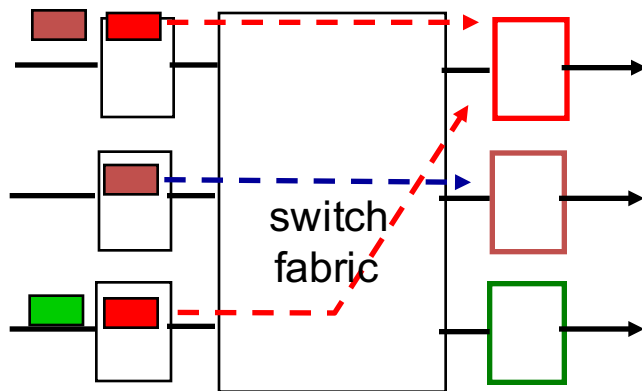
# Contention at input ports

If switching fabric slower than input ports combined

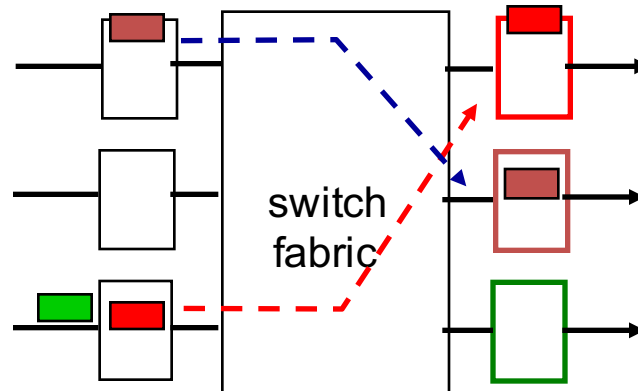
- queueing may occur at input queues
- queueing delay and loss due to input buffer overflow!

Head-of-the-Line (HOL) blocking

- queued pkt at front of queue prevents others from moving forward

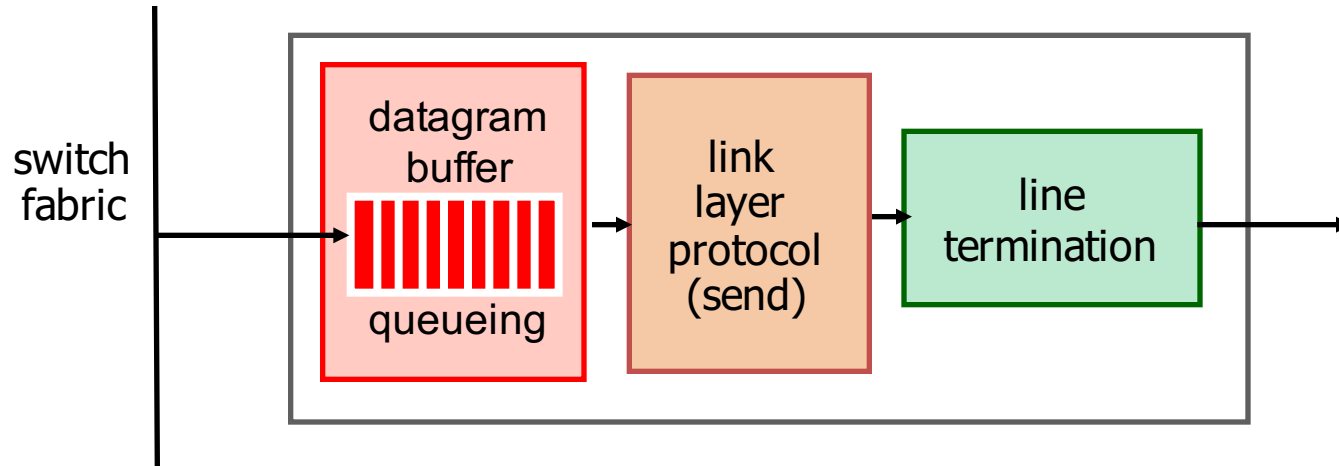


Output port contention: only one red packet can be transferred.  
Lower red packet is blocked



One packet time later: green packet experiences HOL blocking

# Contention at output ports



## Buffering

- when packets arrive from fabric faster than transmission rate
- **packet loss**: due to congestion, lack of buffers

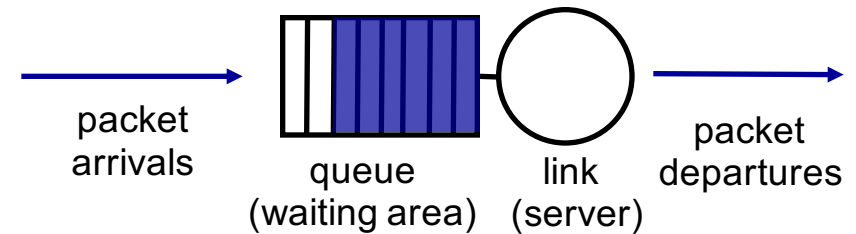
## Scheduling

- chooses next among queued packets to transmit on link
- **net neutrality**: who gets best performance

# Scheduling mechanisms

## FIFO (first in first out)

- send in order of arrival to queue



## Priority

- multiple classes, with different priorities (e.g., based on hdr info)
  - send highest priority queued packet

## Round robin scheduling

- multiple classes, cyclically scan class queues
  - send one packet from each class (if available)

## Weighted fair queueing

- generalized round robin
  - each class gets weighted amount of service in each cycle

In practice: hardware queues use FIFO,  
need software to do priority

**Network Layer**

**INTERNET PROTOCOL**

# Internet Protocol (IP)

## THE network layer protocol of the Internet

- protocol your device **must** implement to run on Internet
- RFC published ~1980

## Provides

- best effort service
  - to get pkts from one end host to another across many interconnected networks using dst IP address in IP hdr
- addressing
  - format and usage of addresses
- fragmentation
  - e.g., if pkt size exceeds Ethernet MTU of 1500 bytes
- some error detection

## Q: what does IP not provide?

- QoS, reliability, ordering, persistent state for e2e flows, connections