

CS120: Computer Networks

Lecture 18. Congestion Control 2

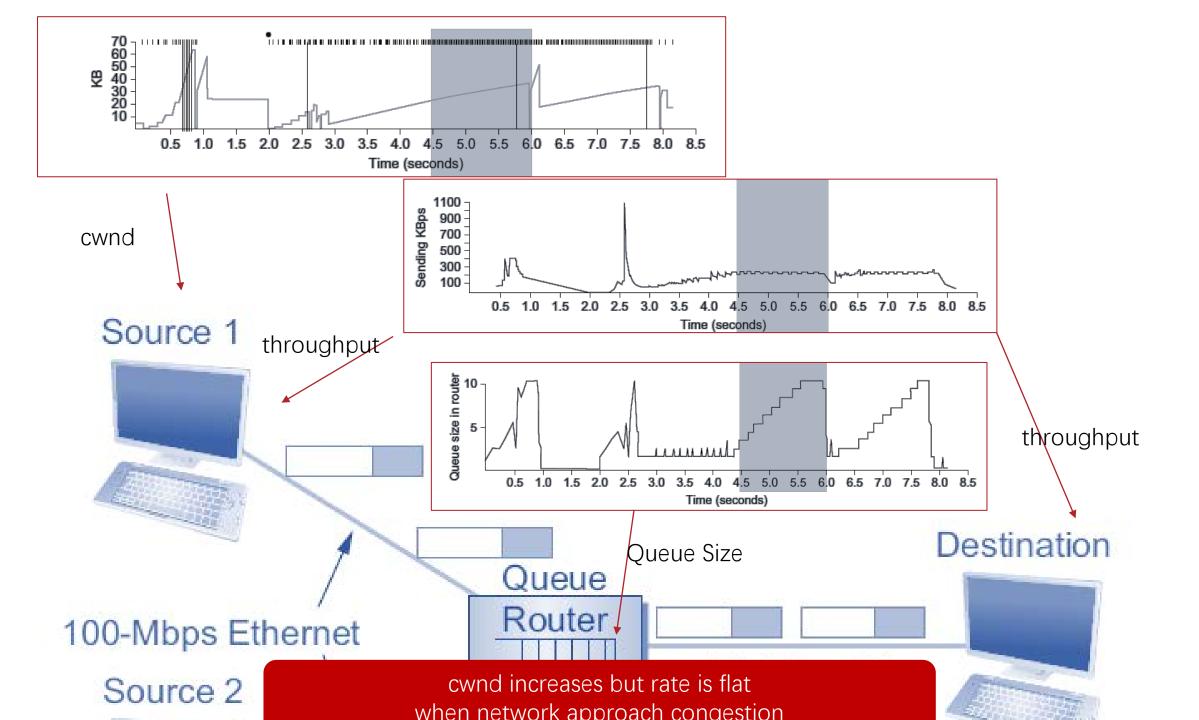
Zhice Yang

Congestion Control

- Host-based Congestion Control
 - Packet Loss
 - **≻**Delay
- Router-based Congestion Control
 - Queuing

Avoid Congestion

 Feedback: Network Delay RTT1 Possible Congestion RTT2 Even No Packet Loss!



TCP Vegas

- Idea: TCP source uses RTT as the sign to avoid network congestion
 - Compare RTT with BaseRTT
- Method:
 - Reduce rate when congestion is about to happen
 - If Actual RTT >> Base RTT
 - Recover rate soon after bandwidth is available
 - if Actual RTT > Base RTT
 - Keep certain pressure on network

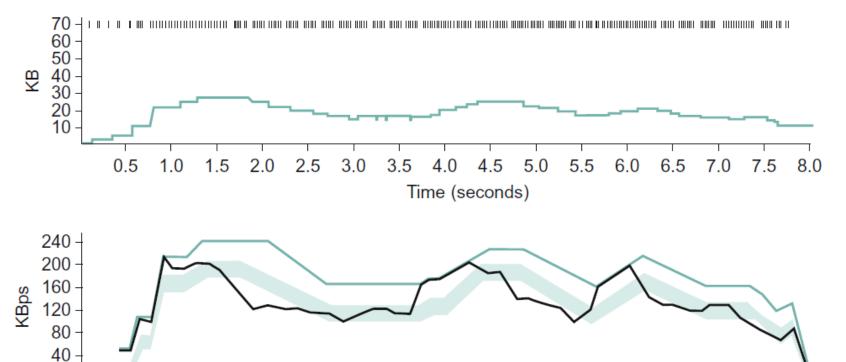
TCP Vegas Algorithm

- BaseRTT: the reference for increases in RTTs
 - The minimum RTT
 - If RTT < BaseRTT
 - BaseRTT = RTT
- ExpectRate:
 - CongestionWindow/BaseRTT
- ActualRate:
 - ActualRTT: according to timestamps
 - ActualRate=CongestionWindow/ActualRTT

TCP Vegas Algorithm

- Source compares ActualRate with ExpectRate
 - if ExpectRate ActualRate < α
 - cwnd++
 - if α < ExpectRate ActualRate < β
 - cwnd = cwnd
 - if β <ExpectRate ActualRate
 - cwnd --

TCP Vegas



Top, congestion window; bottom, expected (colored line) and actual (black line) throughput. The shaded area is the region between the α and β threshold

4.0

4.5

5.0

5.5

6.0

6.5

7.0

7.5

2.5

1.0

1.5

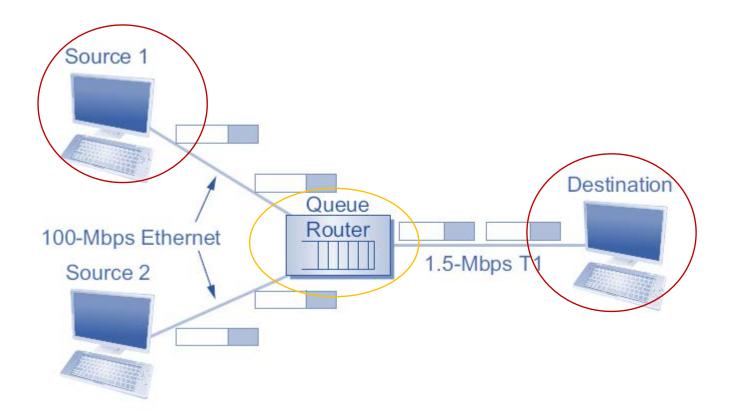
2.0

3.0

3.5

Two Places to Handle Network Congestion

- End hosts
- **≻**Routers

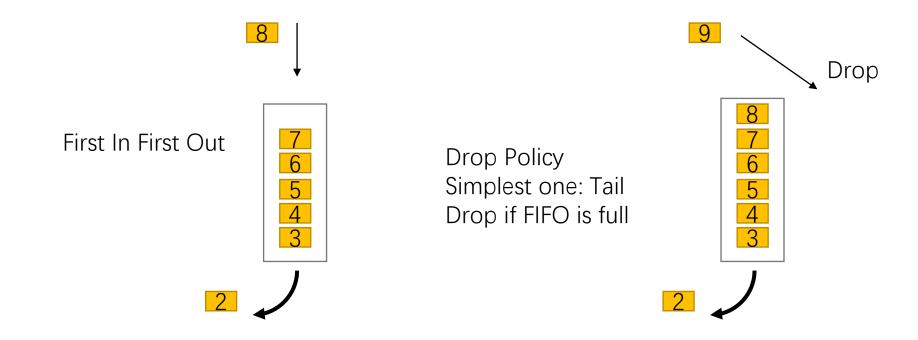


Congestion Control

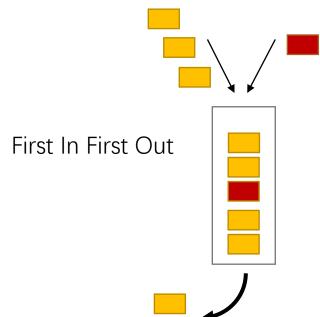
- Host-based Congestion Control
 - Packet Loss
 - Delay
- Router-based Congestion Control
 - **≻**Queuing

- Why ?
 - Queuing discipline in routers determines how packets are transmitted (or dropped)
- Router Resource
 - Bandwidth
 - Which packets get transmitted
 - Queue Buffer
 - Which packets get discarded

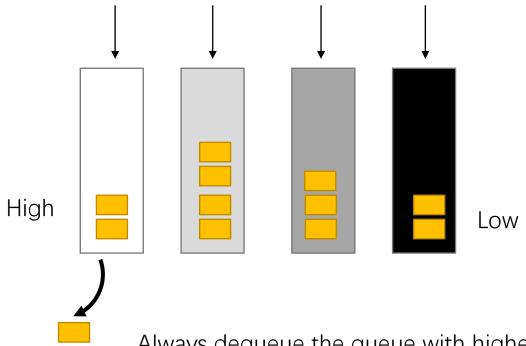
• First-In-First-Out (FIFO)



- Problems in FIFO
 - Too simple to provide resource allocation polices (to avoid congestion)
 - Hard to enforce every network source/flow to follow the same behavior
 - eg. yellow src does not follow congestion control (UDP), can occupy more network source



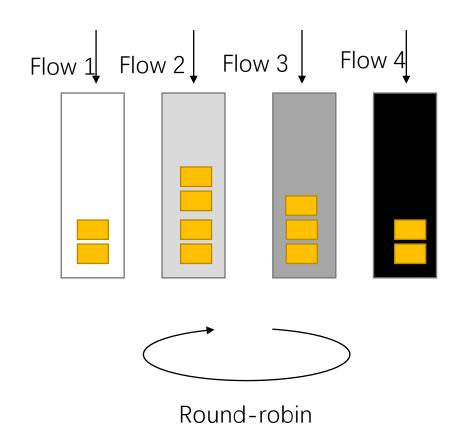
• First-In-First-Out (FIFO) with Priority



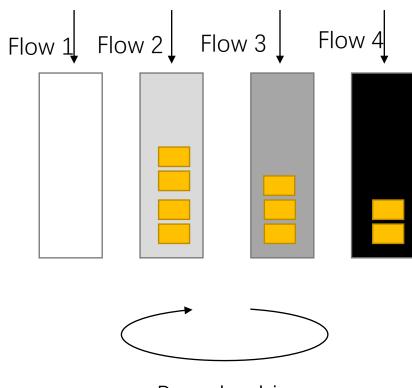
Always dequeue the queue with higher priority unless it is empty

Problem: starvation

- Fair Queuing (FQ)
 - Each flow gets 1/4 output bandwidth

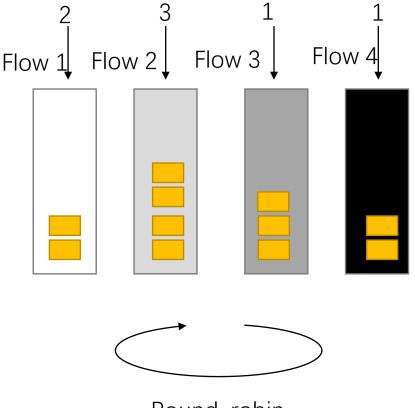


- Fair Queuing (FQ)
 - Each flow gets 1/3 output bandwidth

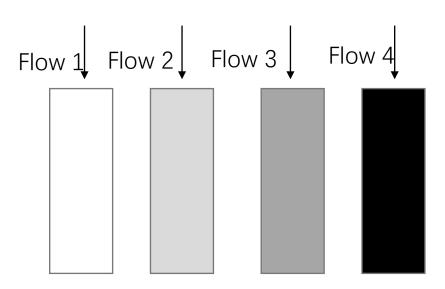


Round-robin

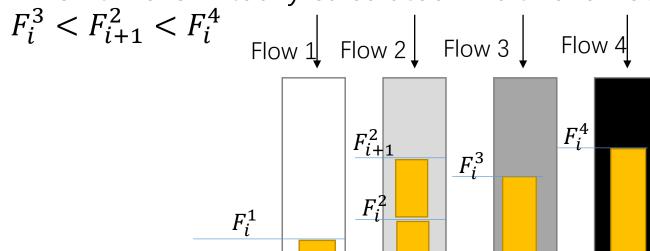
- Weighted Fair Queuing (FQ)
 - Flows with higher weight get more output bandwidth (2/7, 3/7, 1/7, 1/7)



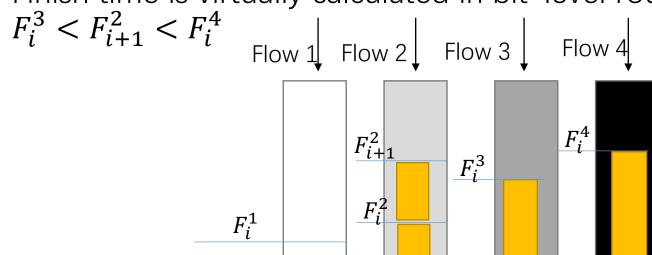
- Bit-Level Fair Queuing (FQ)
 - Schedule according to earliest finish time
 - Finish time of packet i: $F_i = \max(F_{i-1}, A_i) + P_i$
 - P_i is the transmitting duration of packet i, A_i is the arriving time of packet i, F_{i-1} is the finish time of packet i-1 of the same flow



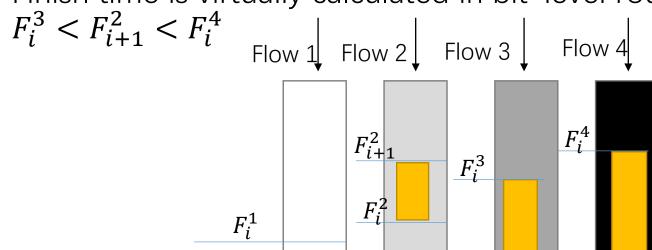
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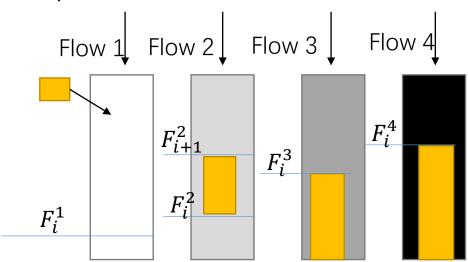
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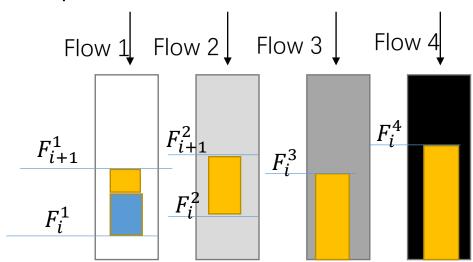
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 - Case: when one queue is idle, idle time also count



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Congestion Control

- Host-based Congestion Control
 - Packet Loss
 - Delay
- Router-based Congestion Control
 - Queuing
 - DECbit
 - Explicit congestion notification (ECN)
 - Random Early Detection (RED)

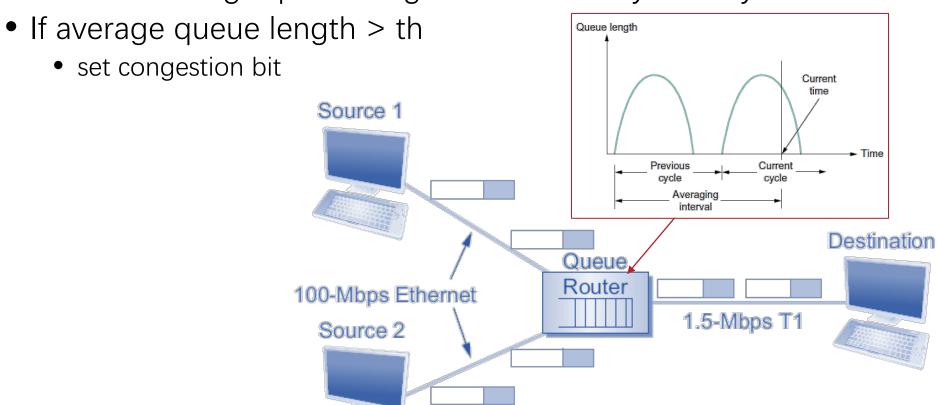
DECbit

- Developed for the Digital Network Architecture (DNA)
 - Before TCP/IP was "standardized"
- Idea: let routers explicitly indicate congestion
- Approach:
 - Router set congestion bit in passing packets if there is congestion
 - Destination echoes bit back to source through acks
 - Source adjusts cwnd according to congestion bit

DECbit

Routers determine congestion

Monitor average queue length over last busy+idle cycle

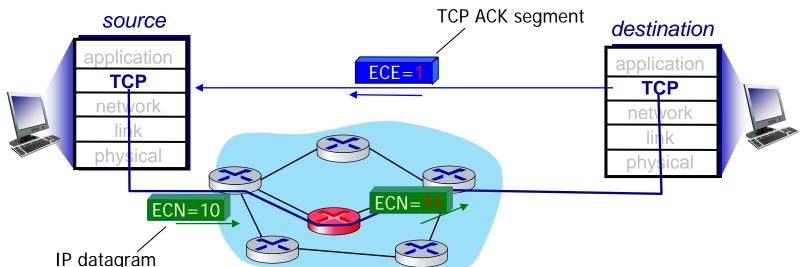


DECbit

- Source reacts to congestion bit
 - If < 50% of last window's packets had bit set
 - cwnd++
 - If > 50% of last window's packets had bit set
 - cwnd*0.875

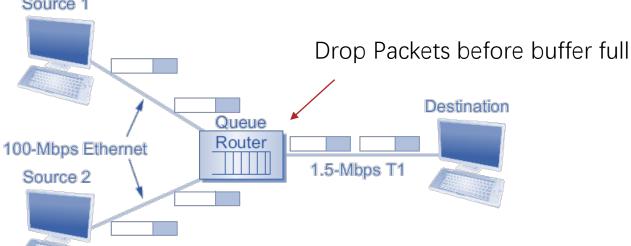
Explicit congestion notification (ECN)

- RFC 3168
 - Two bits in IP header (ECN in IP's ToS field) marked by network router to indicate congestion
 - Mechanism is similar to DECbit
 - Destination sets ECE bit (in TCP Header) on ACK segment to notify sender of congestion



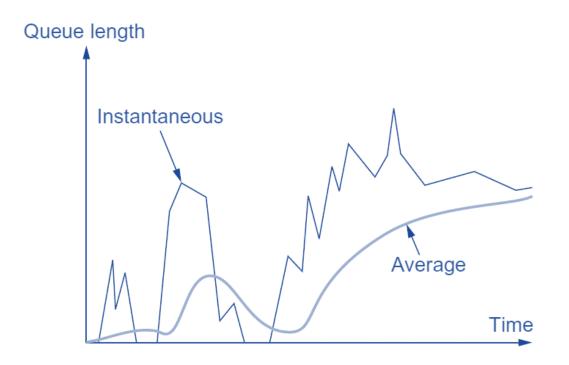
Random Early Detection (RED)

- Another way to offload a part of congestion control function to routers
- Idea: let routers implicitly indicate congestion
 - Router notices that the queue is getting backlogged
 - Router randomly drops packets to signal congestion
 - Source adjusts cwnd



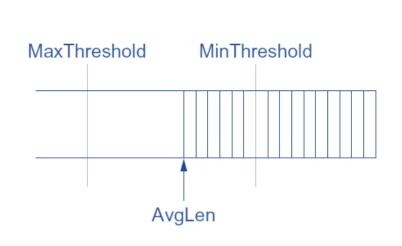
RED Algorithm

- Compute Average Queue Length
 - Moving average
 - AvgLen = (1 Weight) * AvgLen + Weight * SampleLen



RED Algorithm

- Two queue length thresholds
 - if AvgLen <= MinThreshold
 - Enqueue the packet
 - if MinThreshold < AvgLen < MaxThreshold
 - Drop arriving packet with probability P
 - if MaxThreshold <= AvgLen then drop arriving packet





RED Algorithm

- Computing probability P
 - TempP = MaxP * (AvgLen MinThreshold)/(MaxThreshold MinThreshold)
 - P = TempP/(1 count * TempP)
 - Count: number of continuously queued packets without drop
- Why?
 - Drops are spaced out in time
 - Count == 0 => P=TempP
 - Count is large => P=1

Reference

• Textbook 6.4