

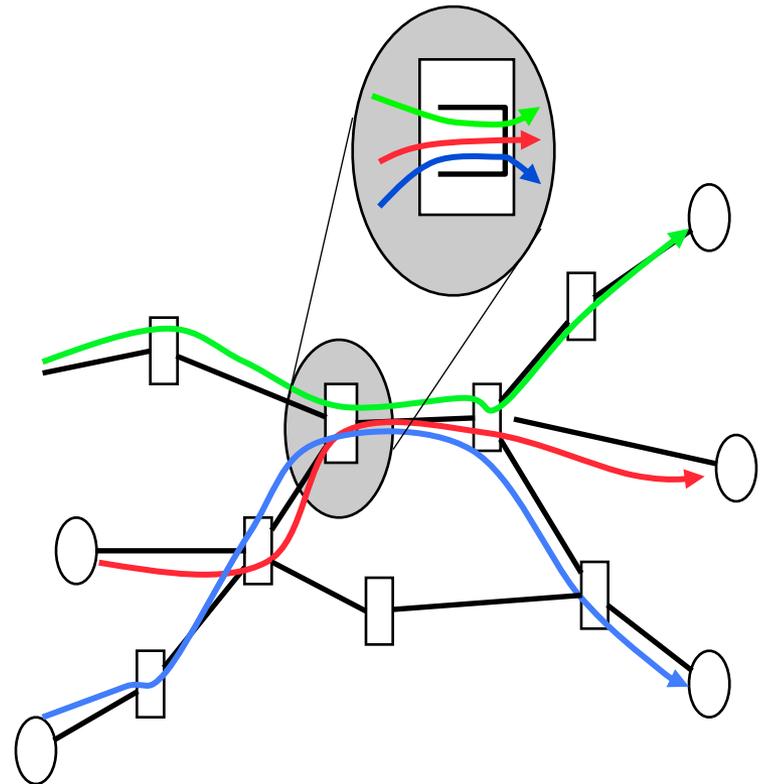
Fair Queueing

Presented by Brighten Godfrey

Slides thanks to Ion Stoica (UC Berkeley)
with slight adaptation by Brighten Godfrey

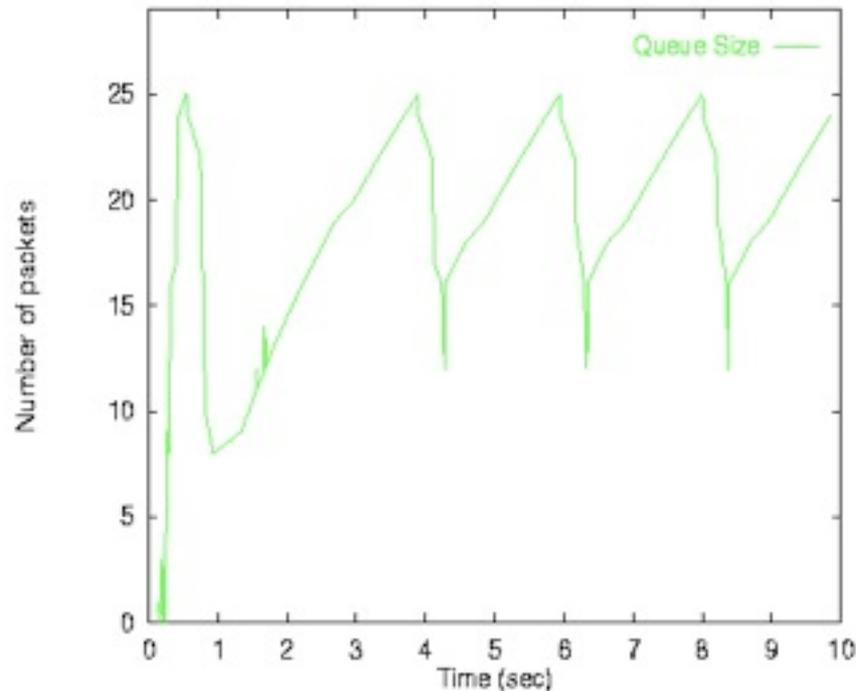
Traditional queueing

- Traditional Internet
 - Congestion control mechanisms at end-systems, mainly implemented in TCP
 - Routers play little role
- Router mechanisms affecting congestion management
 - Scheduling
 - Buffer management
- Traditional routers
 - FIFO
 - Tail drop



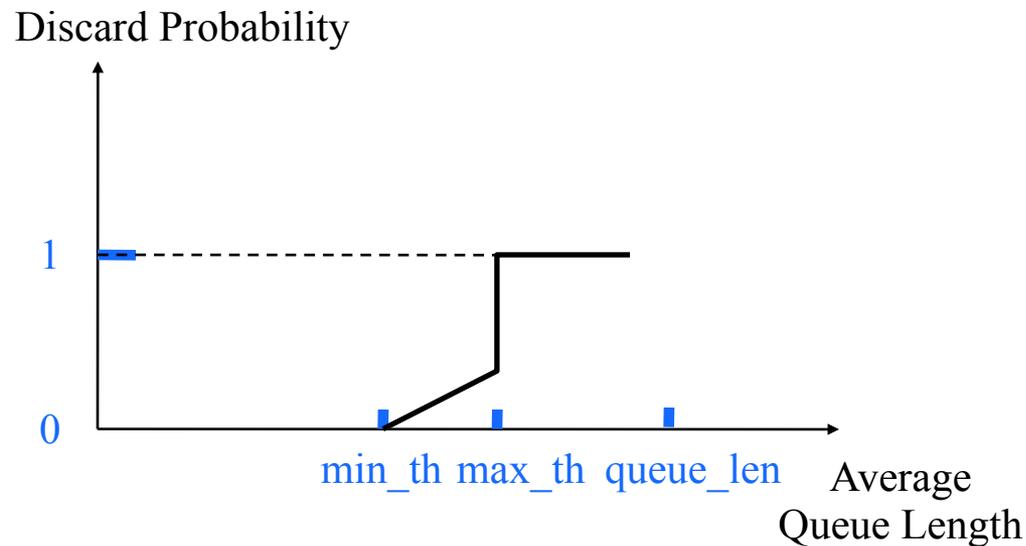
Drawbacks of FIFO with Tail-drop

- Buffer lock out by misbehaving flows
- Synchronizing effect for multiple TCP flows
- Burst or multiple consecutive packet drops
 - Bad for TCP fast recovery



RED

- FIFO scheduling
- Buffer management:
 - Probabilistically discard packets
 - Probability is computed as a function of **average** queue length (why average?)



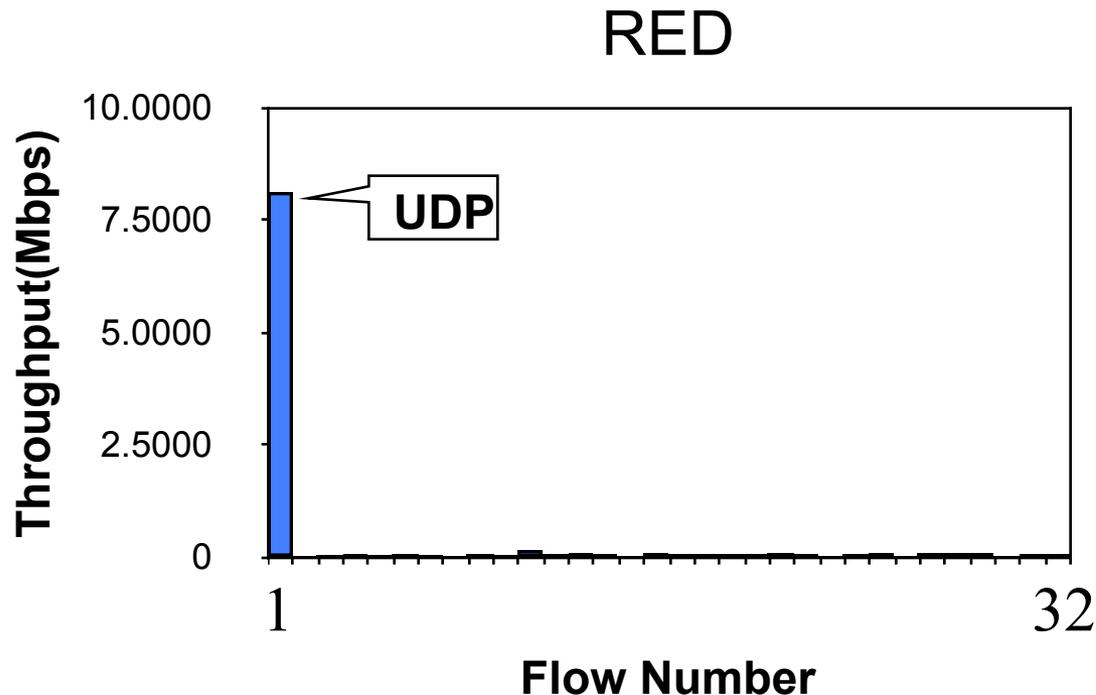
RED Advantages

- Absorb burst better
- Avoids synchronization
- Signal end systems earlier

- And XCP would be even better than RED in these regards

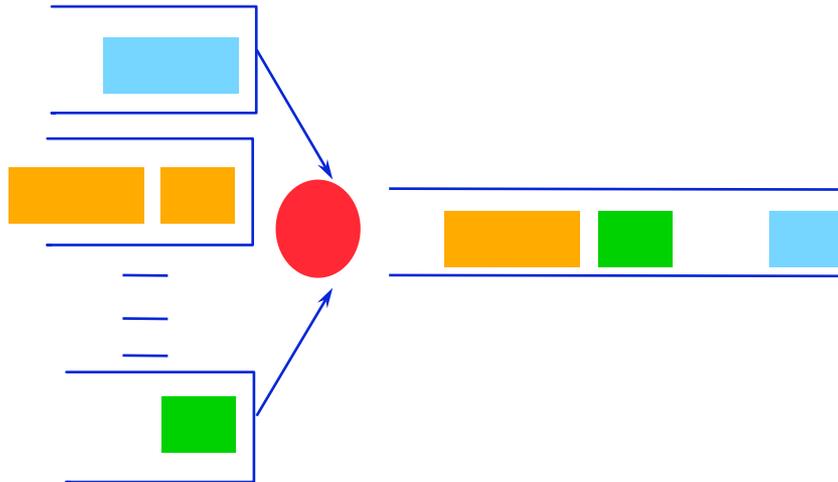
But still no isolation between flows

- No protection: if a flow misbehaves it will hurt the other flows
- Example: 1 UDP (10 Mbps) and 31 TCP's sharing a 10 Mbps link



A first solution

- Round-robin among different flows [Nagle '87]
 - One queue per flow



Round-Robin Discussion

- Advantages: protection among flows
 - Misbehaving flows will not affect the performance of well-behaving flows
 - FIFO does not have such a property
- Disadvantages:
 - More complex than FIFO: per flow queue/state
 - Biased toward large packets – a flow receives service proportional to the number of packets (When is this bad?)

Fair Queueing (FQ) [DKS'89]

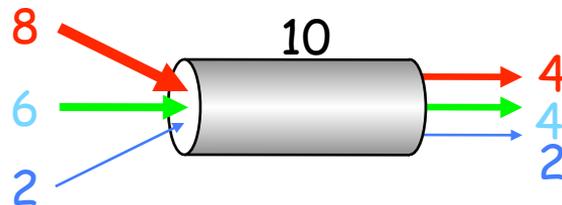
- Define a **fluid flow** system: a system in which flows are served bit-by-bit
 - i.e., **bit-by-bit round robin**
- Advantages
 - Each flow will receive exactly its max-min fair rate
 - ...and exactly its fair per-packet delay

Max-Min Fairness

- Denote
 - C – link capacity
 - N – number of flows
 - r_i – arrival rate
- Max-min fair rate computation:
 1. compute C/N
 2. if there are flows i such that $r_i \leq C/N$, update C and N
$$C = C - \sum_{i \text{ s.t. } r_i \leq C} r_i$$
 3. if no, $f = C/N$; terminate
 4. go to 1
- A flow can receive at most the fair rate, i.e., $\min(f, r_i)$

Example

- $C = 10; r_1 = 8, r_2 = 6, r_3 = 2; N = 3$
- $C/3 = 3.33 \rightarrow C = C - r_3 = 8; N = 2$
- $C/2 = 4; f = 4$

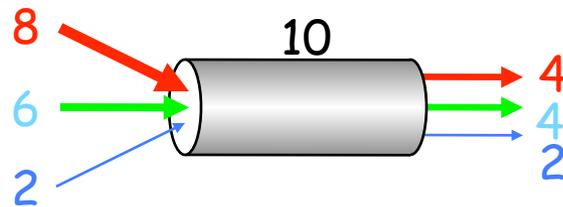


$f = 4:$
$\min(8, 4) = 4$
$\min(6, 4) = 4$
$\min(2, 4) = 2$

Alternate Way to Compute Fair Rate

- If link congested, compute f such that

$$\sum_i \min(r_i, f) = C$$

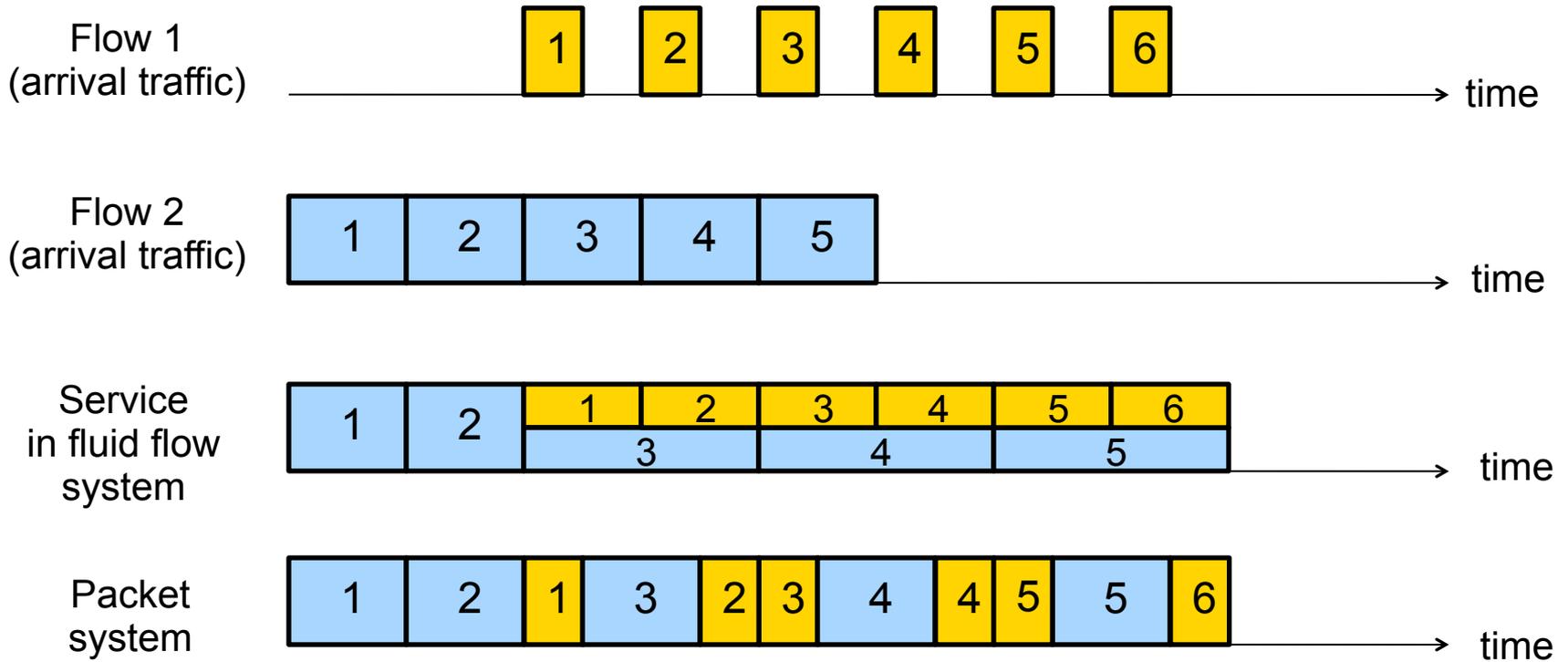


$f = 4:$ $\min(8, 4) = 4$ $\min(6, 4) = 4$ $\min(2, 4) = 2$
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Implementing Fair Queueing

- What we just saw was bit-by-bit round robin
- Can't do it – can't interrupt transfer of a packet (why not?)
- Idea: serve packets in the order in which they would have finished transmission in the fluid flow system
- Strong guarantees
 - Each flow will receive exactly its max-min fair rate (+/- one packet size)
 - ...and exactly its fair per-packet delay (+/- one packet size)

Example



Guarantees

- Translating fluid to discrete packet model doesn't actually involve a lot of combinatorics.
- Theorem: each packet P will finish transmission at or before its finish time in fluid flow model.
 - **assuming** (for now) all packets are in queue at time 0
- Proof:
 - Suppose the packet's finish time is T in fluid model
 - Fluid model: packets that have finished by T sum to $\leq RT$ bits (possibly less: some packets may still be in progress) where R is link rate
 - Packet model: these will be sent in time $\leq RT / R = T$.
- So, why is the real guarantee (without assumption) only approximate (\pm one packet)?

Problem

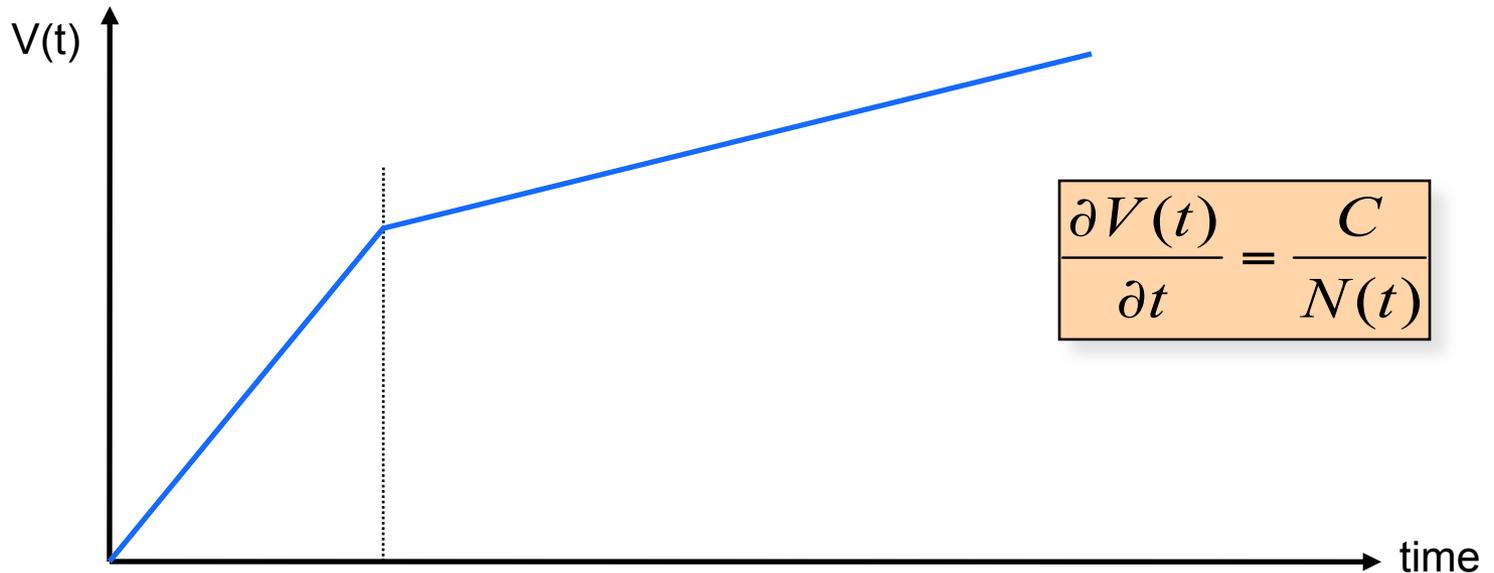
- Recall algorithm: “serve packets in the order in which they would have finished transmission in the fluid flow system”
- So, need to compute finish time of each packet in the fluid flow system
- ... but new packet arrival can change finish times of packets in the system (perhaps all packets!)
- Updating those times would be expensive

Solution: Virtual Time

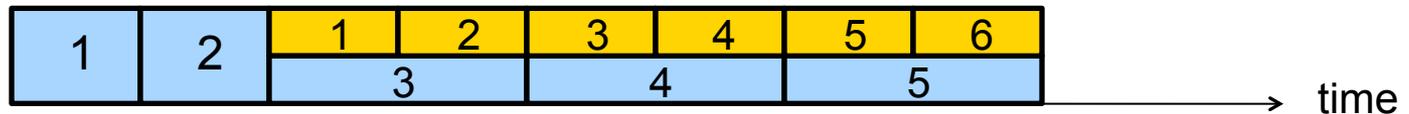
- Key Observation: while the finish times of packets may change when a new packet arrives, the order in which packets finish doesn't!
 - Only the order is important for scheduling
- Solution: instead of the packet finish time maintain the number of rounds needed to send the remaining bits of the packet (**virtual finishing time**)
 - Virtual finishing time doesn't change upon packet arrival
- System virtual time – index of the round in the bit-by-bit round robin scheme

System Virtual Time: $V(t)$

- Measure service, instead of time
- $V(t)$ slope – rate at which every active flow receives service
 - C – link capacity
 - $N(t)$ – number of active flows in fluid flow system at time t



Service
in fluid flow
system



Fair Queueing Implementation

- Define
 - F_i^k - finishing time of packet k of flow i (in system virtual time reference system)
 - a_i^k - arrival time of packet k of flow i
 - L_i^k - length of packet k of flow i

- Virtual finishing time of packet $k+1$ of flow i is

$$F_i^{k+1} = \max(V(a_i^k), F_i^k) + L_i^{k+1}$$

- Order packets by increasing virtual finishing time, and send them in that order

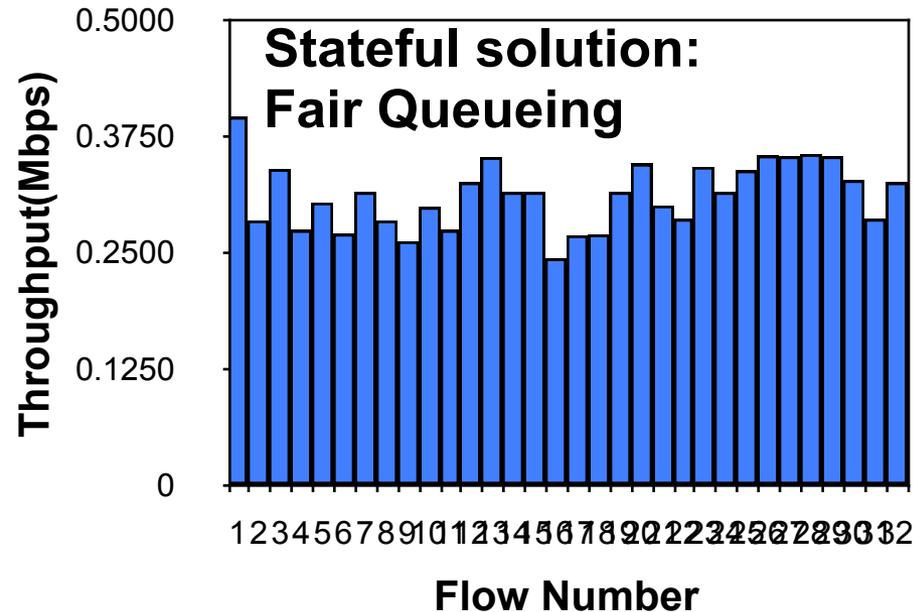
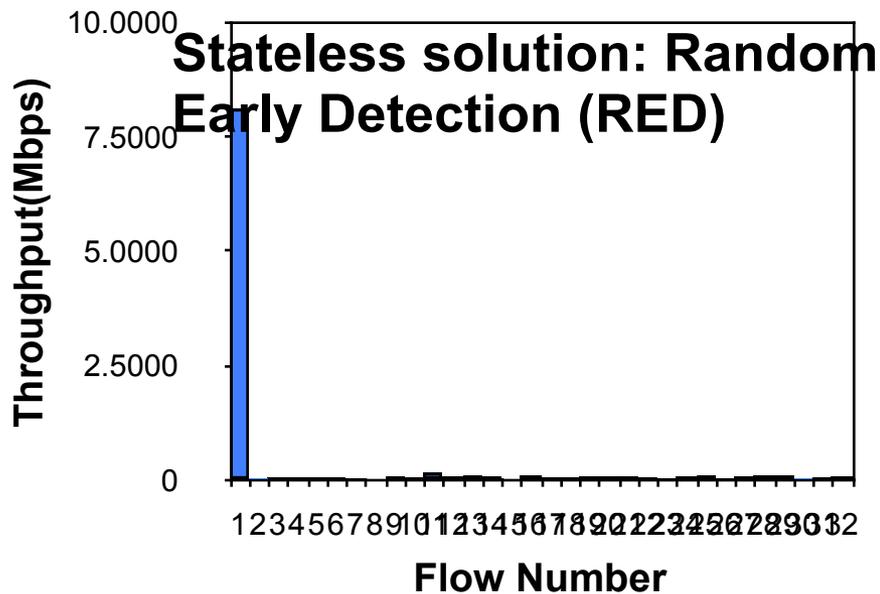
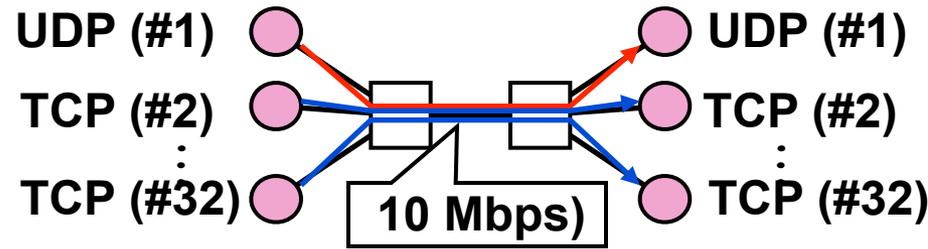
“Weighted Fair Queueing” (WFQ)

- What if we don't want exact fairness?
 - E.g.,: file servers
- Assign weight w_i to each flow i
- And change virtual finishing time

$$F_i^{k+1} = \max(V(a_i^k), F_i^k) + \frac{L_i^{k+1}}{w_i}$$

Simulation Example

- 1 UDP (10 Mbps) and 31 TCPs sharing a 10 Mbps link



Summary

- FQ does not eliminate congestion; it just manages the congestion
- You need both end-host congestion control and router support for congestion control
 - End-host congestion control to adapt
 - Router congestion control to protect/isolate
- Don't forget buffer management: you still need to drop in case of congestion. Which packet's would you drop in FQ?
 - One possibility: packet from the longest queue

Announcements

- Got my emails?
- Project proposals due Tuesday
- Watch for survey