

Advanced Computer Networks

Transport in Data Center Networks (III)

<https://pages.cs.wisc.edu/~mgliu/CS740/F25/index.html>

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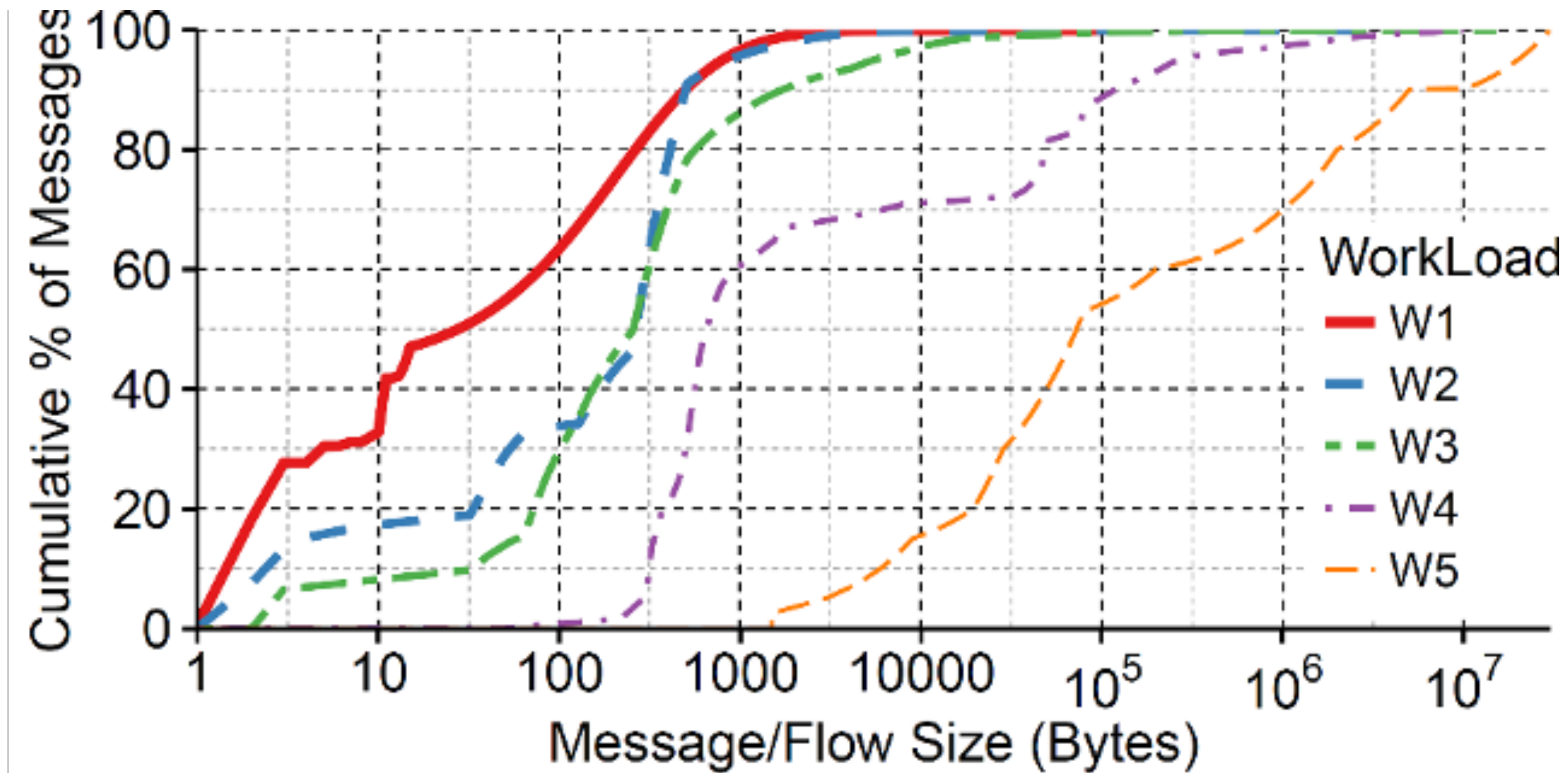
Outline

- Last lecture
 - Transport in Data Center Networks (II)
- Today
 - Transport in Data Center Networks (III)
- Announcements
 - Midterm report due today 11:59 PM
 - Lab2 due 11/05/2025 11:59 PM

Problem: How can we achieve low latency of tiny messages under high networking load?

Short Messages Dominate Workloads

- W1, W2: 95% of messages shorter than 1000 bytes



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3X!

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Near-hardware tail latency is hard!

Observation: get rid of any queueing effect

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Shallow queue \neq Zero queue

Can we remove the queue?

~~**Can we remove the queue?**~~

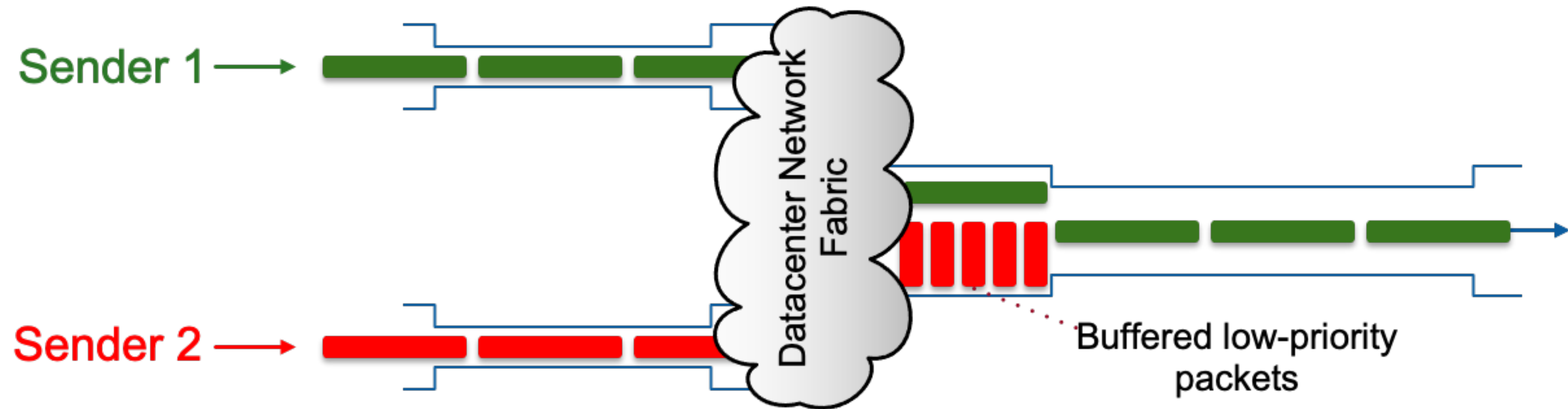
Transit Burst

High BW Utilization

How about bypassing the queue?

Networking Priorities

- Use hardware-provided priority queues to reduce queueing delay



The key idea of Homa is using priorities effectively to bypass buffering.

How does Homa work?

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- TCP: three-way handshake
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 - Optimistic and assume there will be enough capacity
- **Homa: no connection setup**
 - **Similar assumptions as NDP**
 - **Connection-less message-based protocol**

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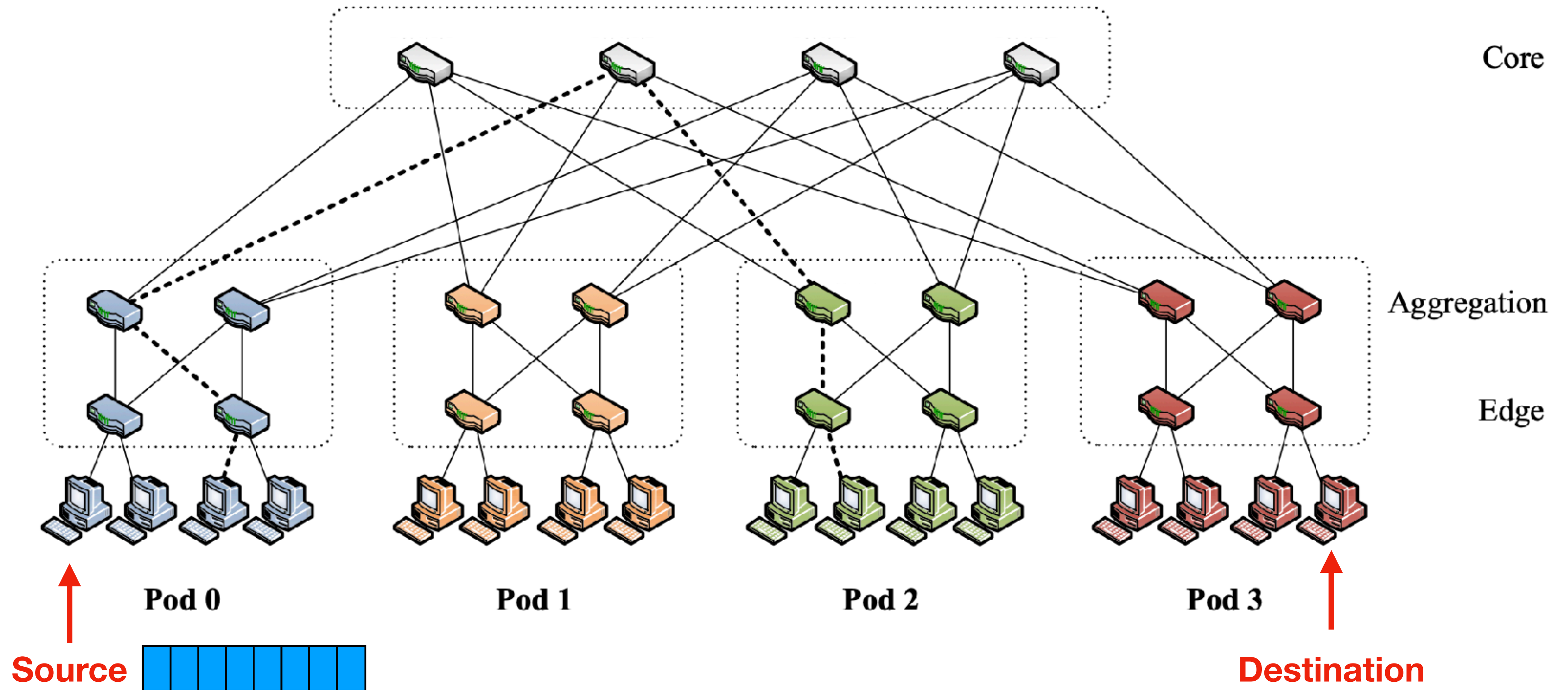
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- **Homa: full bandwidth-delay product (BDP)**
 - **Unscheduled packets**

#3: Routing @Switch

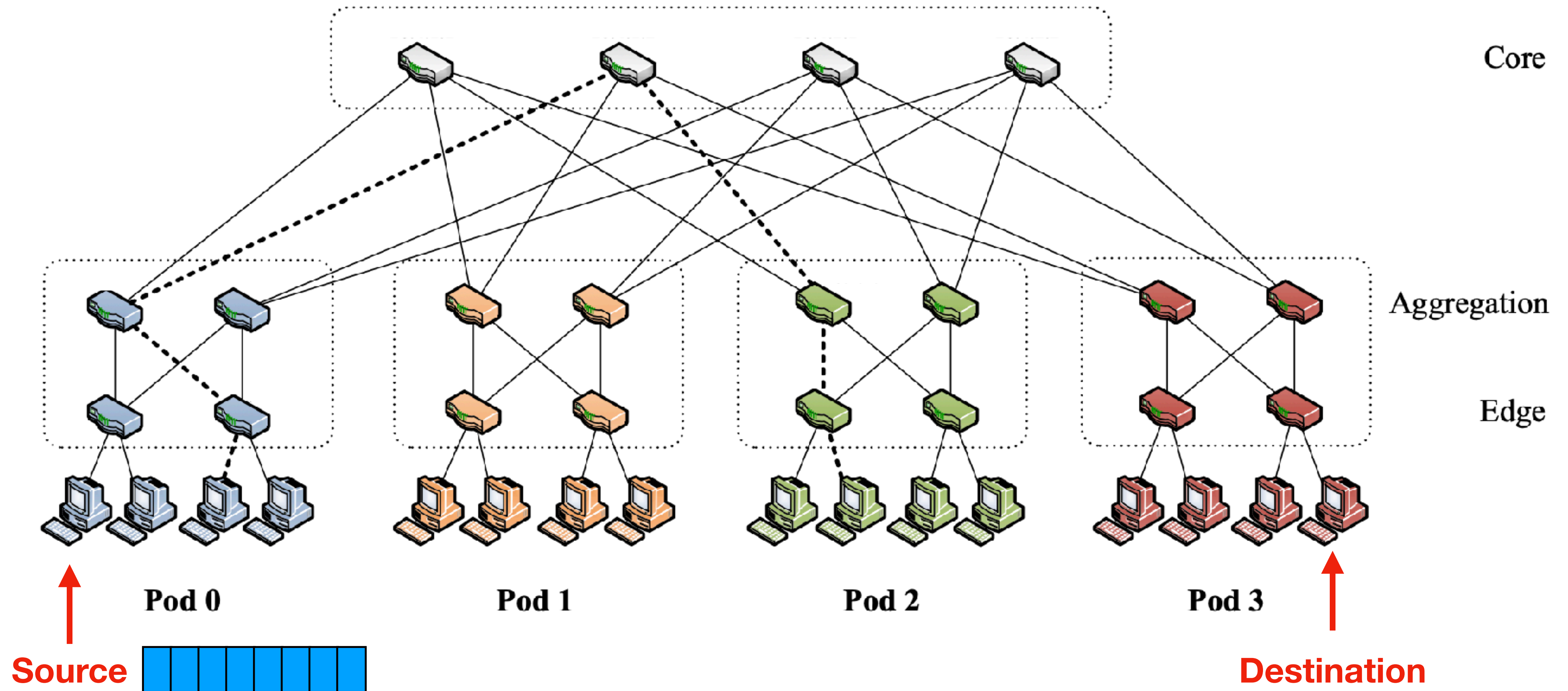
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- TCP: per-flow routing



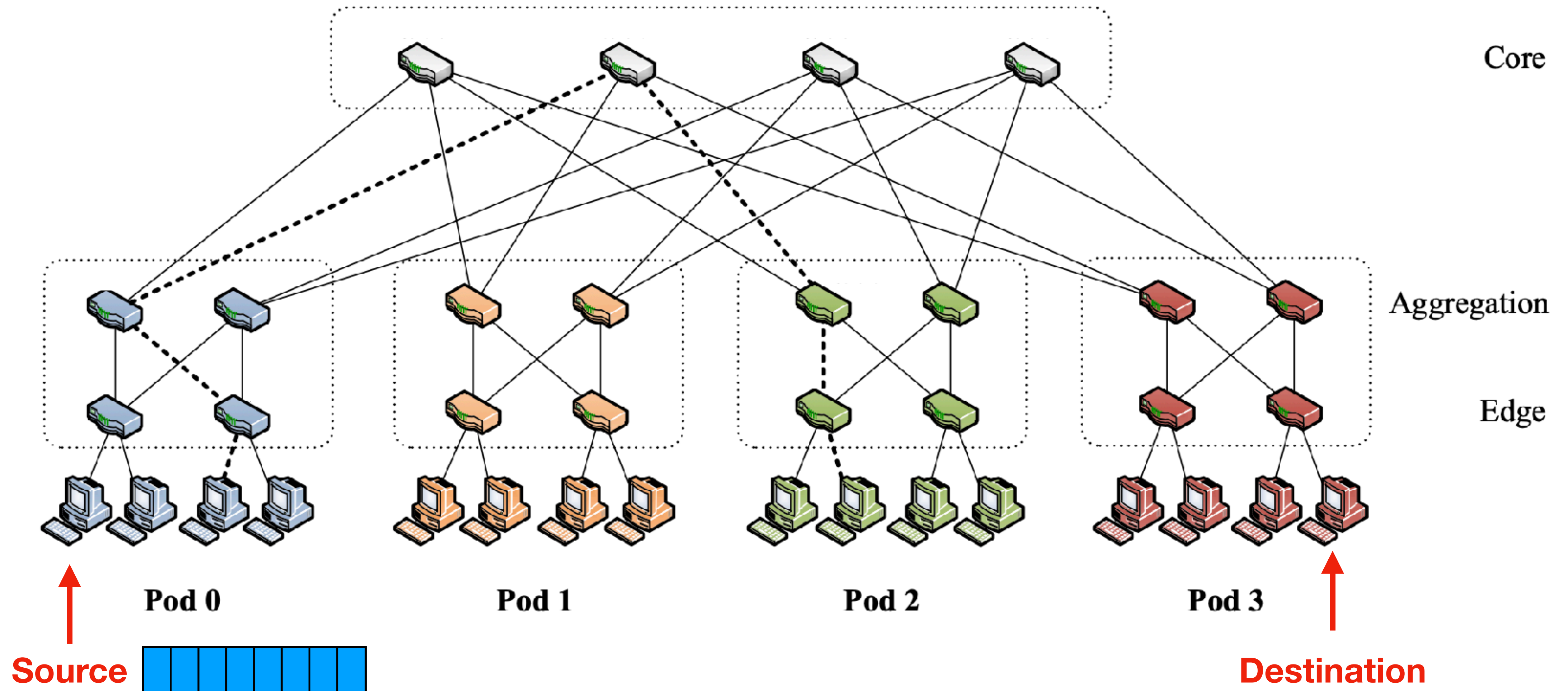
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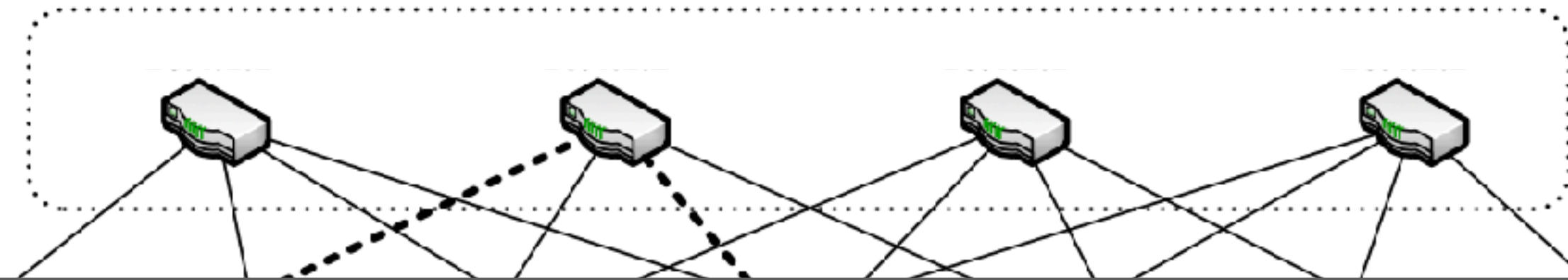
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Core

- Homa allows per-packet ECMP depending on the packet header and how the ECMP hash is calculated



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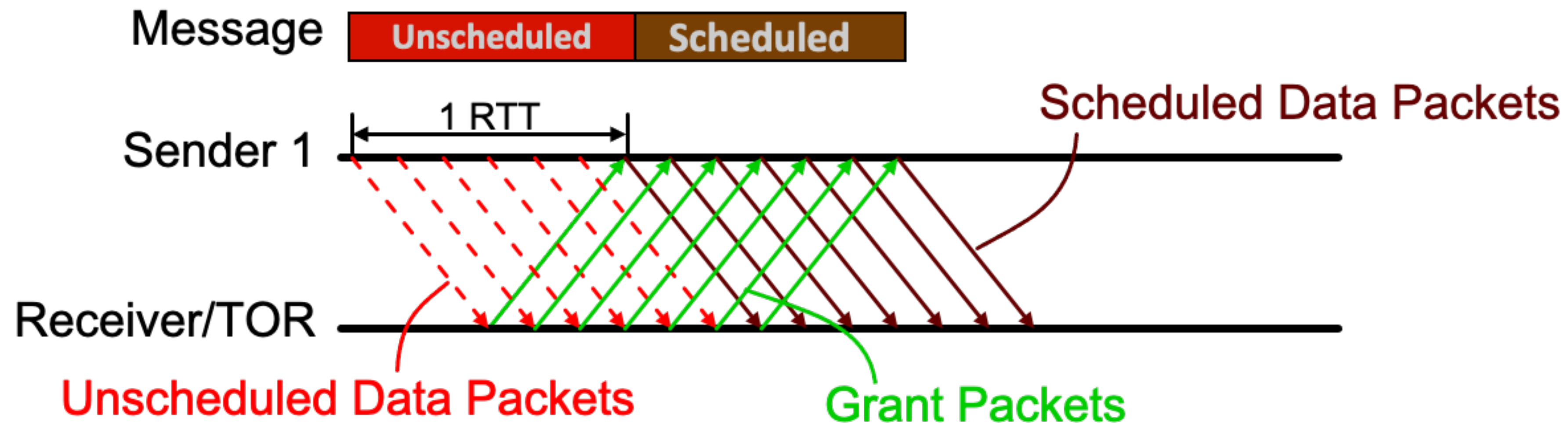
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- **Homa: Receiver-driven**
 - **Network priority kicks in**

Homa Receiver-driven Transport

- The receiver determines how to schedule the remaining flow
 - Unscheduled -> Scheduled
 - Priority: label each packet based on the urgency
 - Grant: provide transmission permission to the sender, one per packet



Homa Receiver-driven Transport

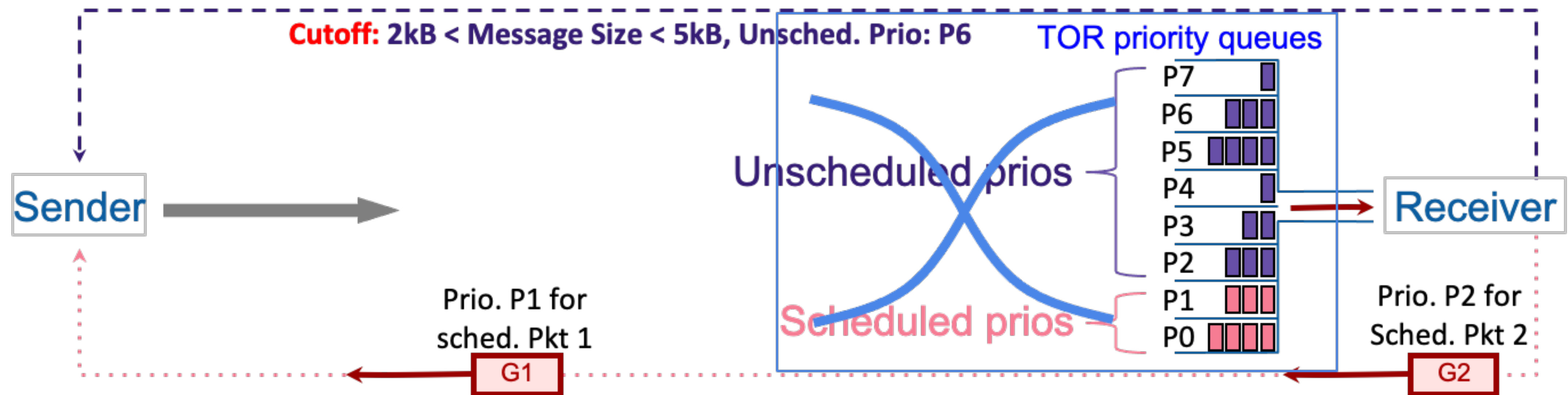
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- How does Homa assign priorities?
- How many grants does a receiver allocate?



Dynamic Priority Assignment

- Principles
 - Unscheduled packets: pre-assigned with higher priorities
 - Scheduled packets: adaptive priority specified in each grant



Dynamic Priority Assignment

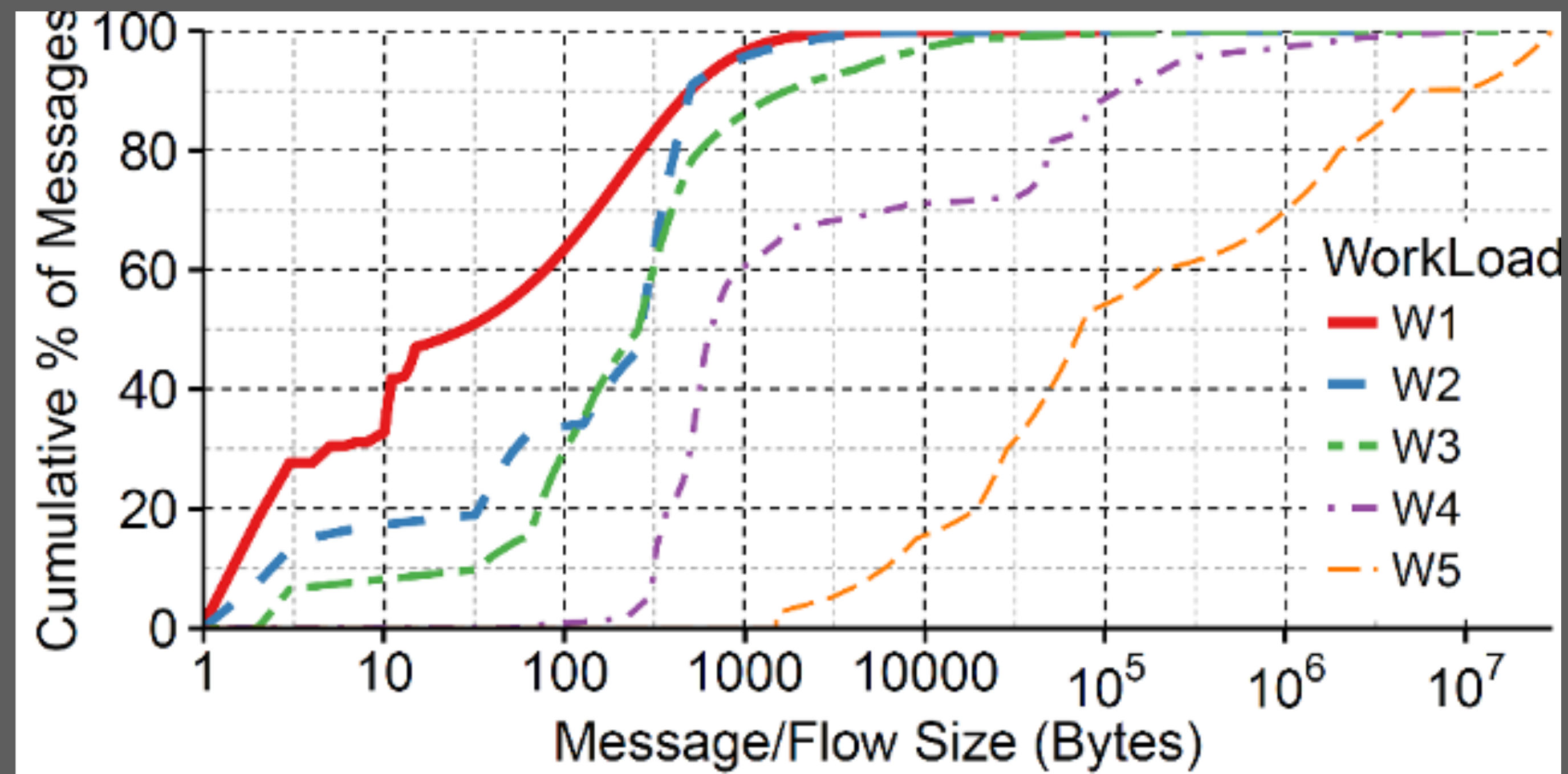
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Why unscheduled packets receive higher priority?

Dynamic Priority Assignment

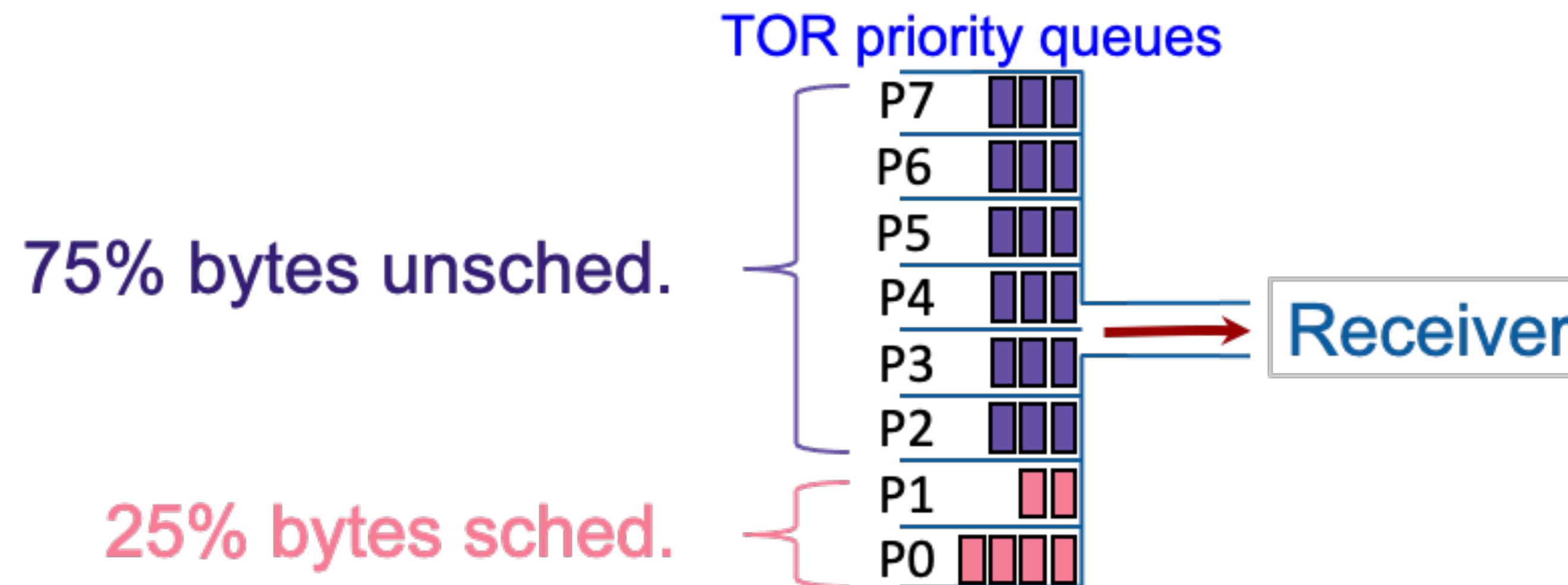
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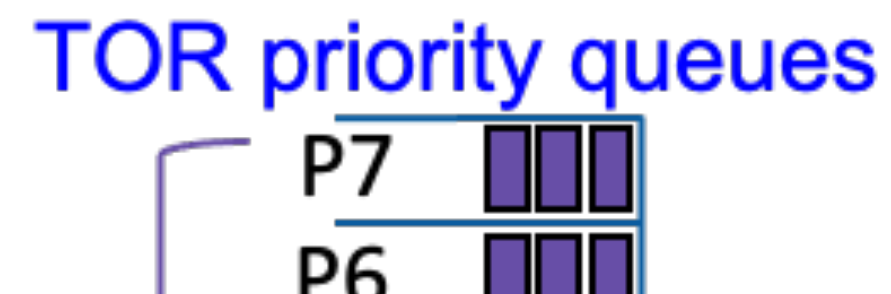
Priority Assignment for **Unscheduled Packets**

- Pick cut-offs based on CDF of message sizes
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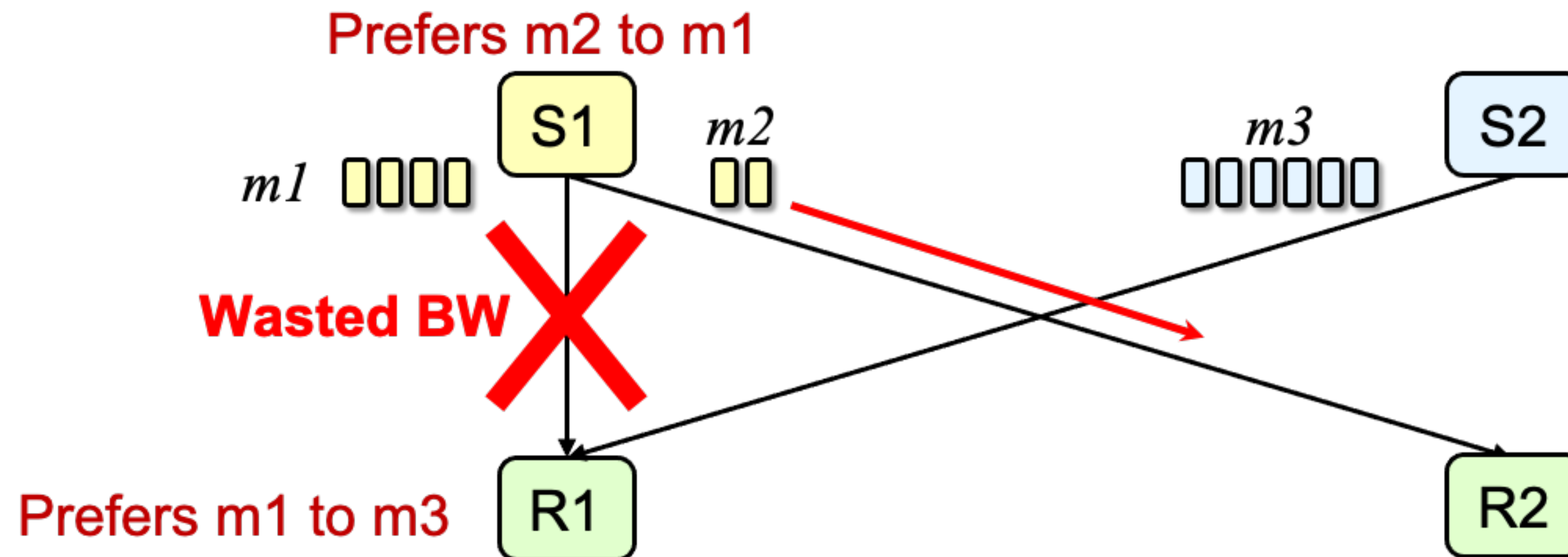
Fraction of priorities for unscheduled packets =
Fraction of incoming bytes unscheduled

Priority Assignment for **Scheduled Packets**

- Allocate adaptively based on incoming messages
 - Start with the lowest priority
 - Use higher priority for pad preemption

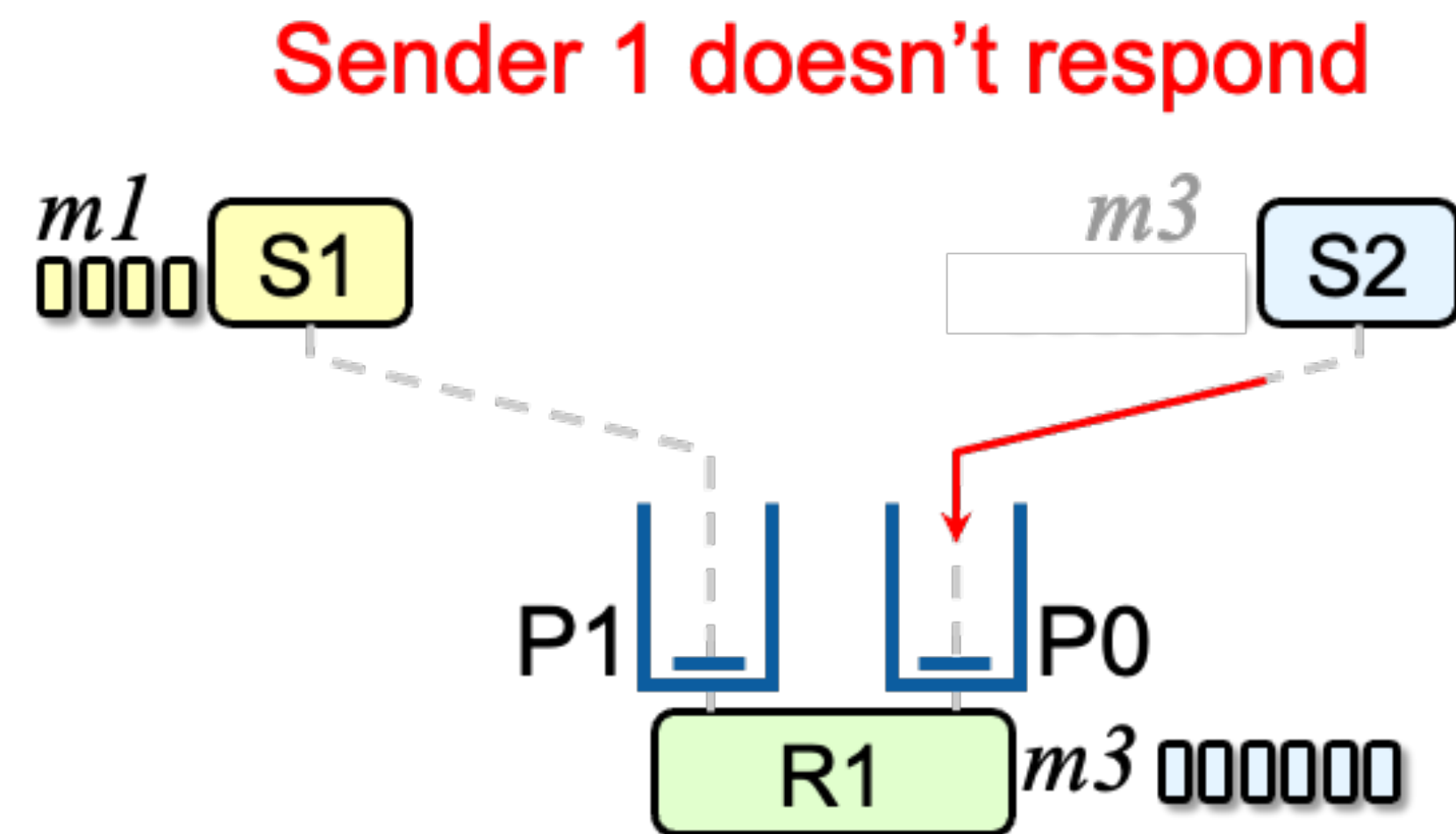
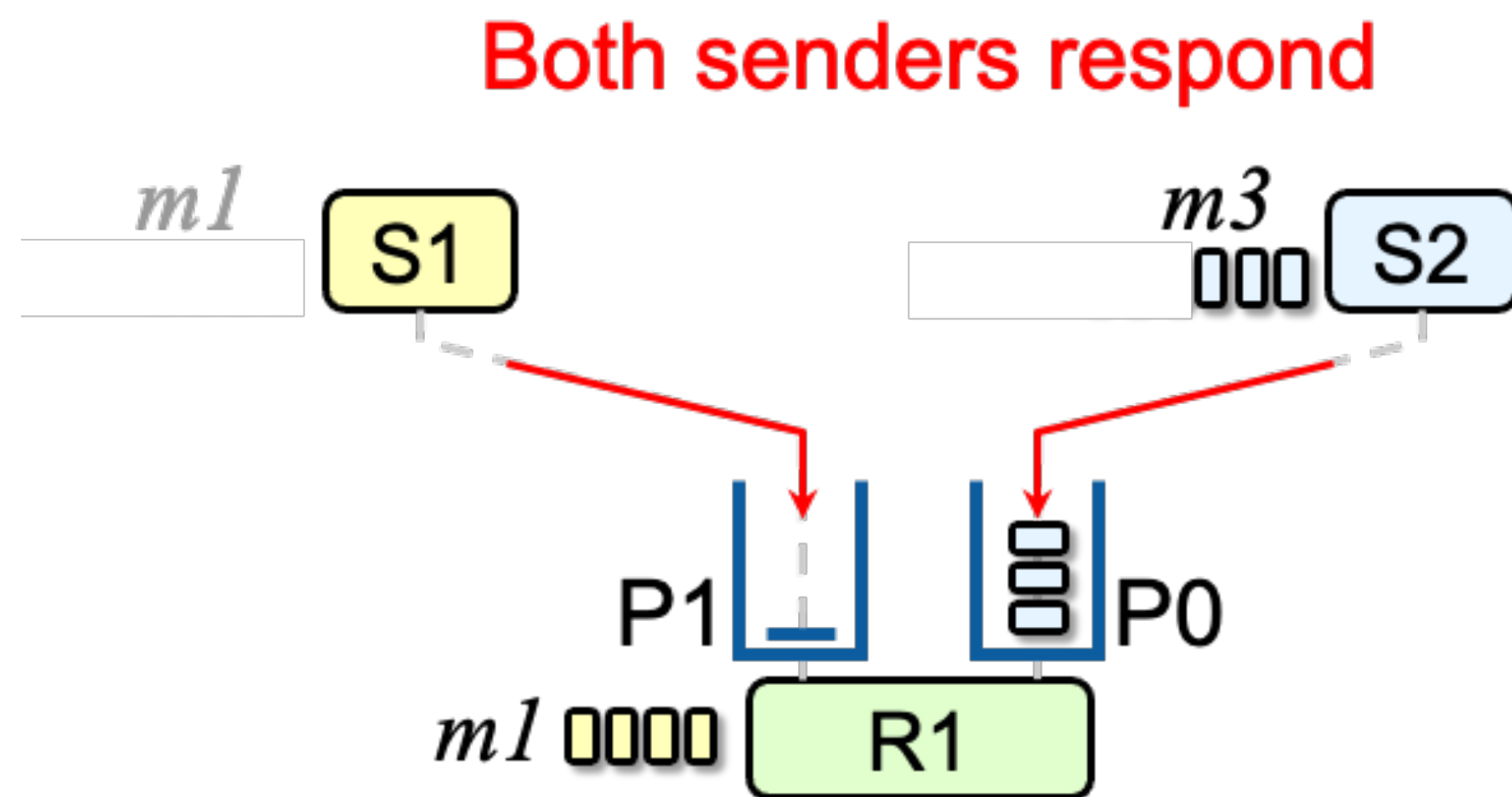
An issue: Priority Mechanisms Hurt Utilization

- Senders may not respond promptly to grant



Controlled Overcommitment

- Use priorities to favor short message
- Use buffering to achieve high bandwidth usage



How many grants are issued?

Use Grants for Rate Control

- Tell the sender to send N remaining bytes of data
- $N = \text{Min} (\text{BDP}, \text{Remaining bytes \#}, \text{Offset})$
 - Offset, determined based on receiving buffer availability and fairness

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- **Homa: receiver-driven**
 - **Grant packets are used for pacing**
 - **Priority is used for minimizing the impact of large flows**

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- **Homa: N/A**
 - **The sender marks the last packet**
 - **Grants can still be received**

What happens if the switch queue is full?

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- NDP: packet trimming
 - Headers of trimmed packets are used for traffic control
 - Co-design the switch behavior with the transport protocol
- **Homa: drop**
 - **The switch performs priority packet scheduling**

What happens if packets are delivered out-of-order?

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- NDP: common behavior due to per-packet routing
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 - Maintain a separate pull sequence space for each connection
- **Homa: common behavior**
 - **Use priority to escalate the scheduled packet (re)transmission**

What happens if links or switches fail?

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- **Homa: timeout engineering**
 - **Little discussion**

Is Homa fair?

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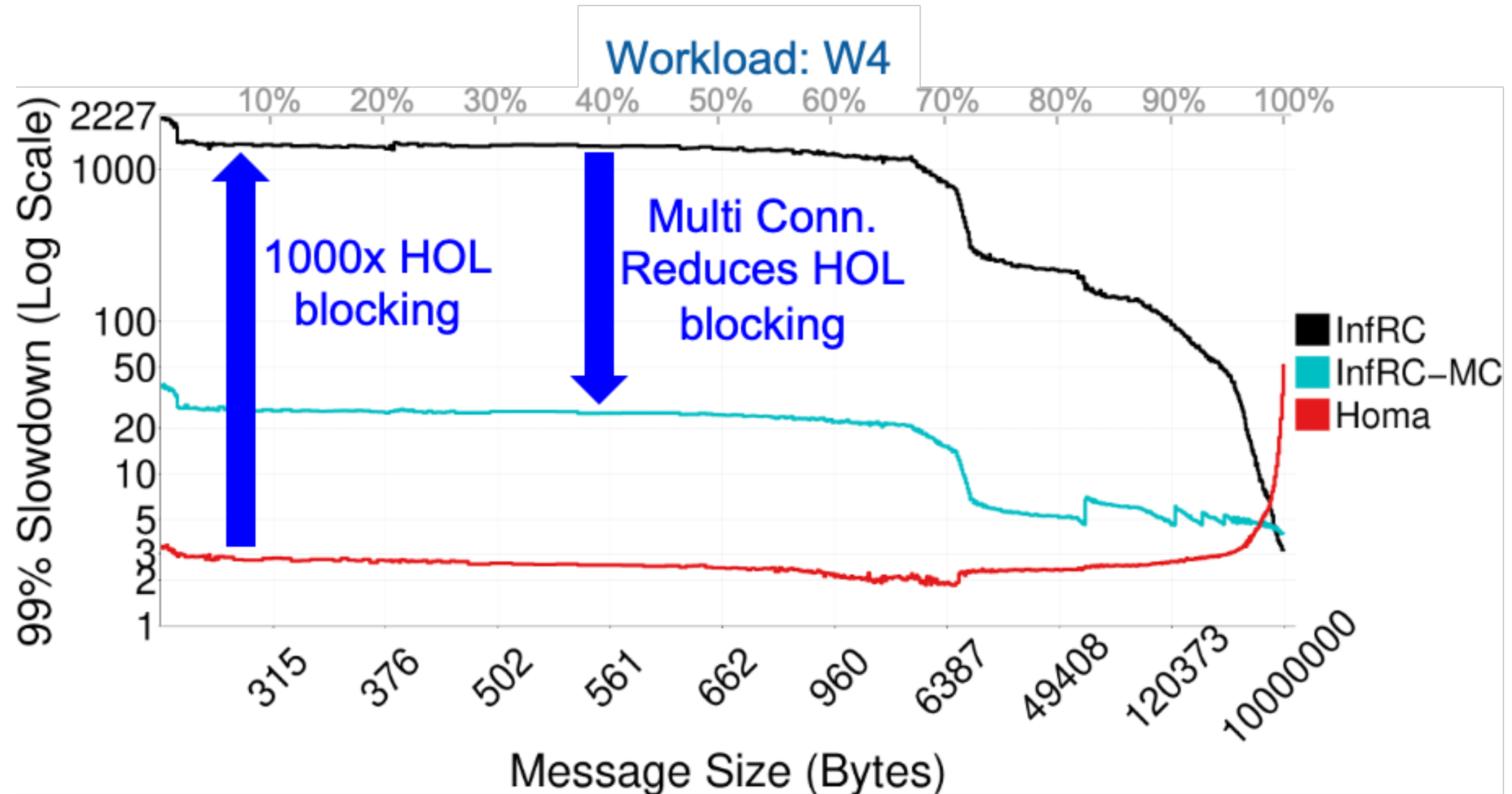
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- **Homa**: fairness depends on priority
 - Small flows send unscheduled packets based on the message CDF
 - Large flows send scheduled packets based on priority adaptivity

Homa Evaluation

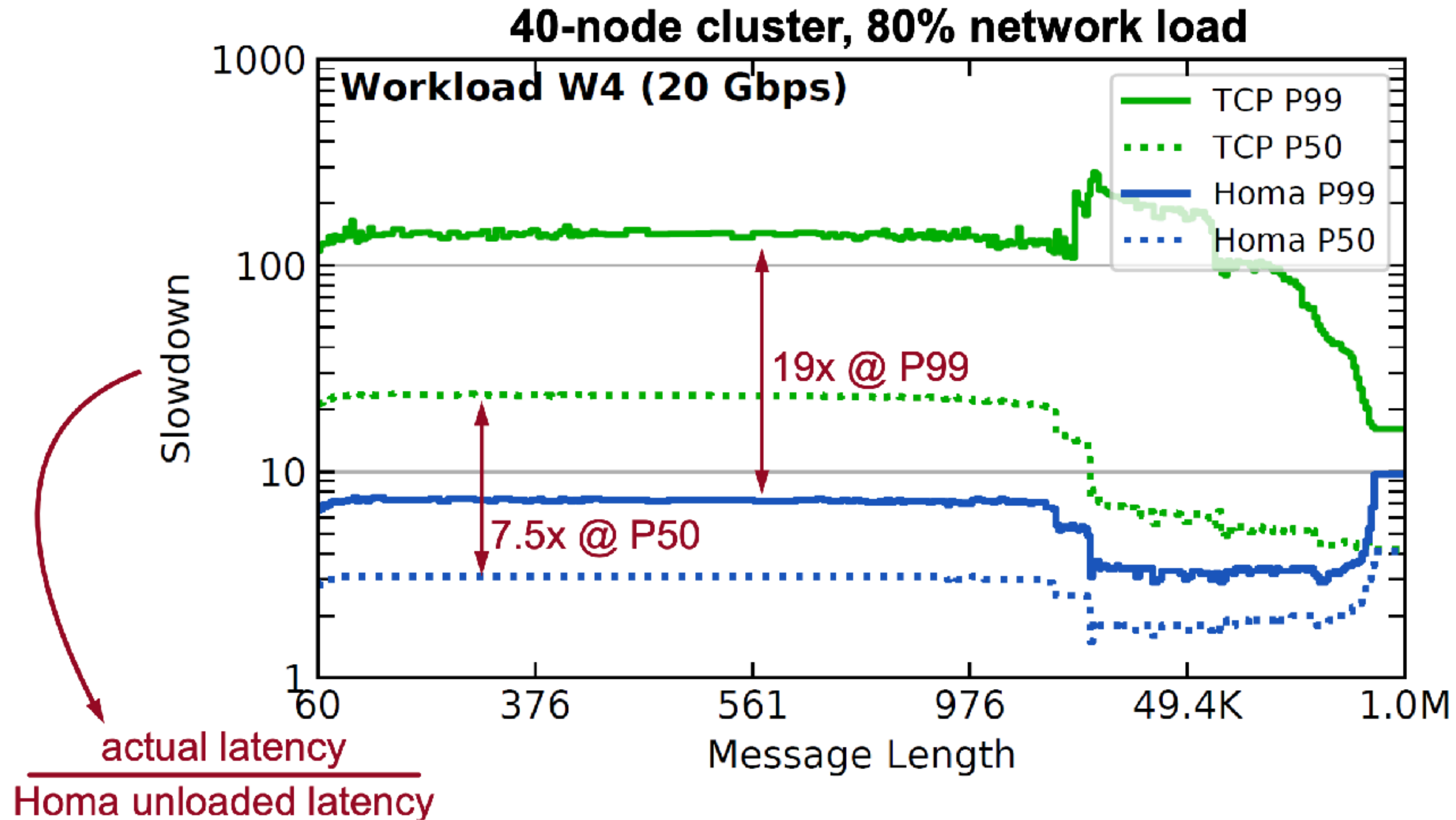


Homa/Linux (ATC'21)

- Target: RPC-based applications in data centers
 - Message-oriented
 - Connection-less

```
int homa_send(int sockfd, const void *request, size_t reqlen,  
              const struct sockaddr *dest_addr, socklen_t addrlen,  
              uint64_t *id);  
  
int homa_reply(int sockfd, const void *response, size_t resplen,  
               const struct sockaddr *dest_addr, socklen_t addrlen,  
               uint64_t id);  
  
int homa_recv(int sockfd, void *buf, size_t len, int flags,  
              struct sockaddr *src_addr, socklen_t addrlen,  
              uint64_t *id);
```

Homa/Linux >> TCP/Linux



Still Have Some Overheads

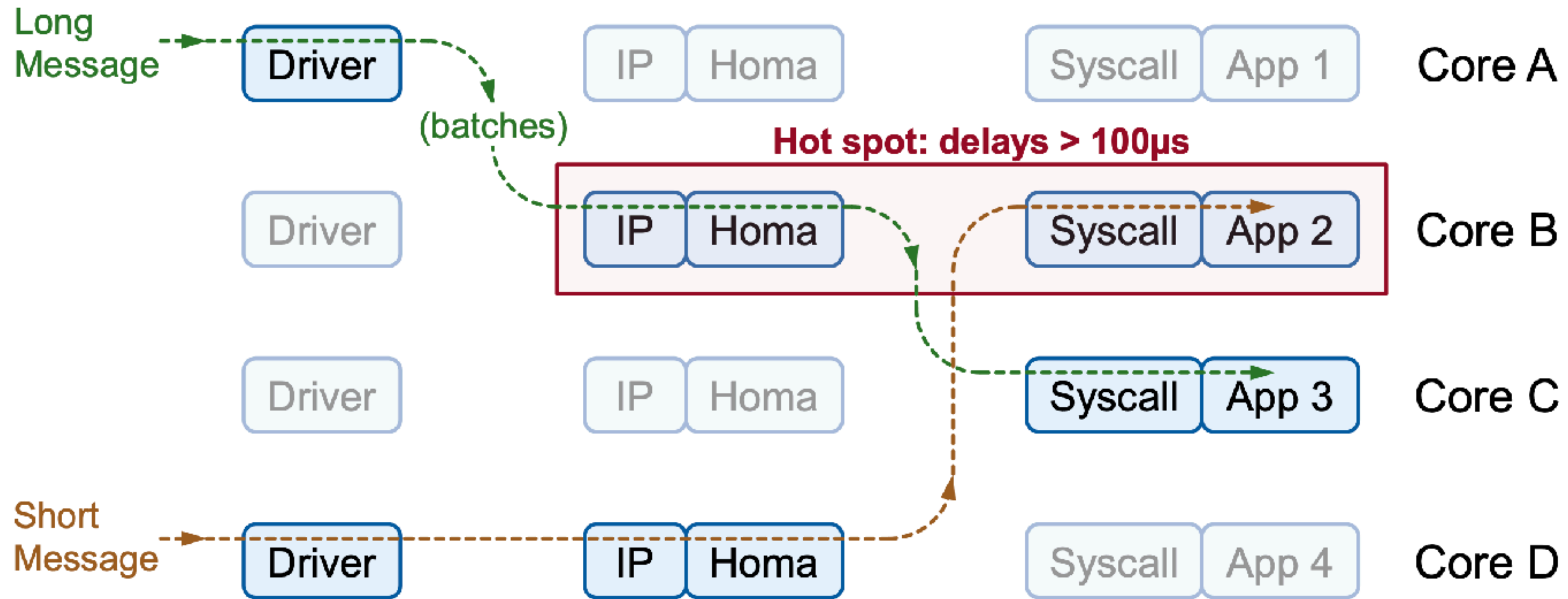
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Load balancing is hard! — the path between NIC to cores

Load Balancing Causes Hot Spots



Homa/Linux v.s. Snap (SOSP'19)

- Google's user-space nstack implementation

	Homa	Snap
Base latency (polling)	15.1 μ s	9 μ s
Cores to drive 80 Gbps bidirectional	17	7–14

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But still suffers from load-balancing issues. For example, throughput per core drops by 3.5x - 7x.

Homa v.s. TCP

- **Connection oriented**
 - High time/space overheads (datacenter apps have 1000's of connections)
- **Stream oriented**
 - Awkward for RPCs (transport doesn't know message boundaries)
 - Head-of-line blocking
- **Fair sharing of bandwidth**
 - Increases latency, especially for short messages
- **Sender-driven congestion control**
 - Requires buffer occupancy to detect congestion
 - Buffer occupancy → high latency
- **Requires in-order packet delivery**
 - Cripples load balancing

<https://arxiv.org/abs/2210.00714>

Summary

- Today
 - Homa
- Next topic: Endhost Networking Stack
 - Linux NStack (Sigcomm'21)
 - SNAP (SOSP'19)