Midterm Review

CPSC 433/533, Spring 2021 Anurag Khandelwal

- Test will be during class hours, starting at 1.00pm, ending at 1:15pm (a total of **75** minutes)
 - Will be Zoom proctored: join via this zoom link, and keep your videos on!
 - We will keep track of Zoom participants and compare to submissions.
 - Please aim to join the Zoom meeting 5 minutes early!

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 - Please aim to join the Zoom meeting 5 minutes early!
- Closed book, closed notes, etc.
- You can prepare a I page cheat-sheet, handwritten notes only
 - No calculators (test does not require any complicated calculations)

- Test only assumes material covered in lecture
 - Read text to clarify details and context for the above

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 - Use this as a hint to check whether you are on the right track
- You do need to understand how and why things work!
 - Understand pros/cons, when a solution is applicable/useful/useless, etc.
- You do need to understand enough to design new protocols
 - Use the expertise you have gained with my incessant questioning during class!

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 - Two parts: 20 points + 40 points

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Second Part: Longer questions (how & why things work)

• "I was having a nice day, minding my own business, and then BGP went nuts..."

Cheating: Don't do it

- You will be required to sign a honor code at the beginning of the exam
- Questions are conceptual require thinking, will take time
- Don't waste it trying to cheat; we have ways of finding out

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- My plan: summarize, not explain
 - Stop me when you want to discuss something further!

Topics

- Basic concepts (Lecture 1, 2)
- Architecture & Principles (Lecture 3, 4)
- Network Layer (Lectures 4-10)
 - Concepts: valid routing state, convergence, least-cost paths
 - Overall context (inter- and intra-domain routing)
 - Computing least-cost routes (DV, LS)
 - IP addressing
 - Inter-domain routing
 - Router architecture
- Transport (Lectures 10-14)
 - Role of the transport layer
 - UDP vs.TCP
 - TCP basics: reliability
 - TCP flow control
 - TCP congestion control
 - TCP critiques & router assistance

Basic Concepts

You should know:

- Statistical multiplexing
- Packet vs. circuit switching
- Link characteristics
- Packet delays

How are network resources shared?

Two approaches

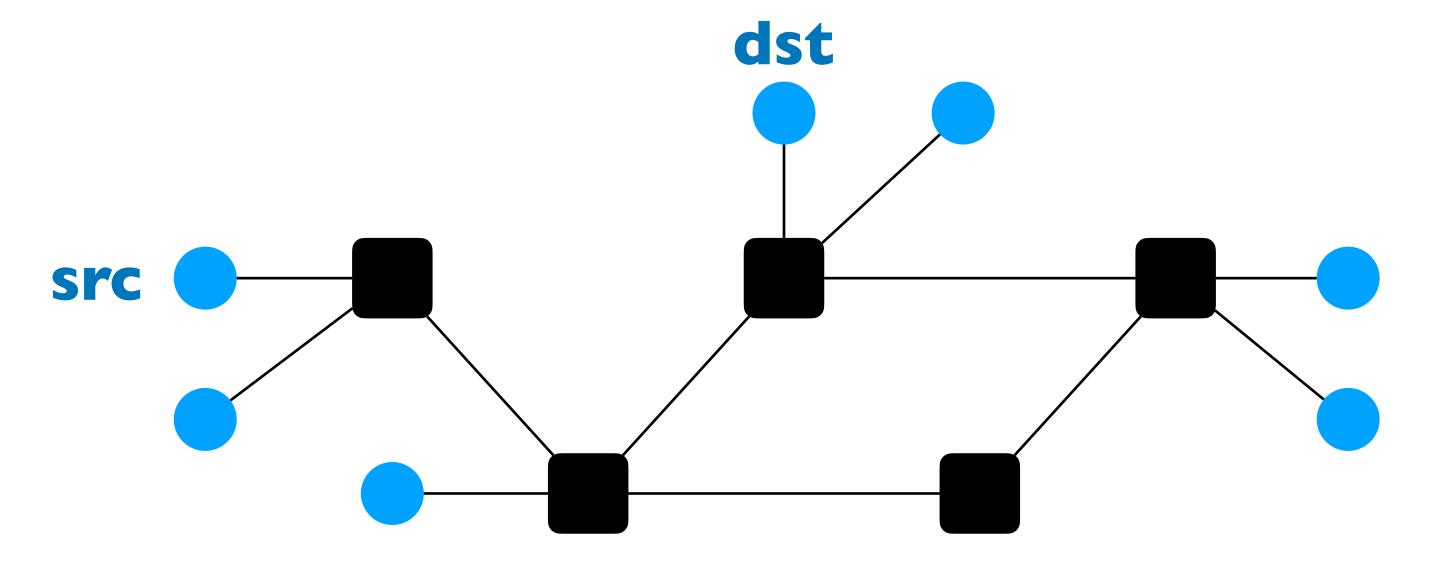
- Reservations → circuit switching
- On demand → packet switching

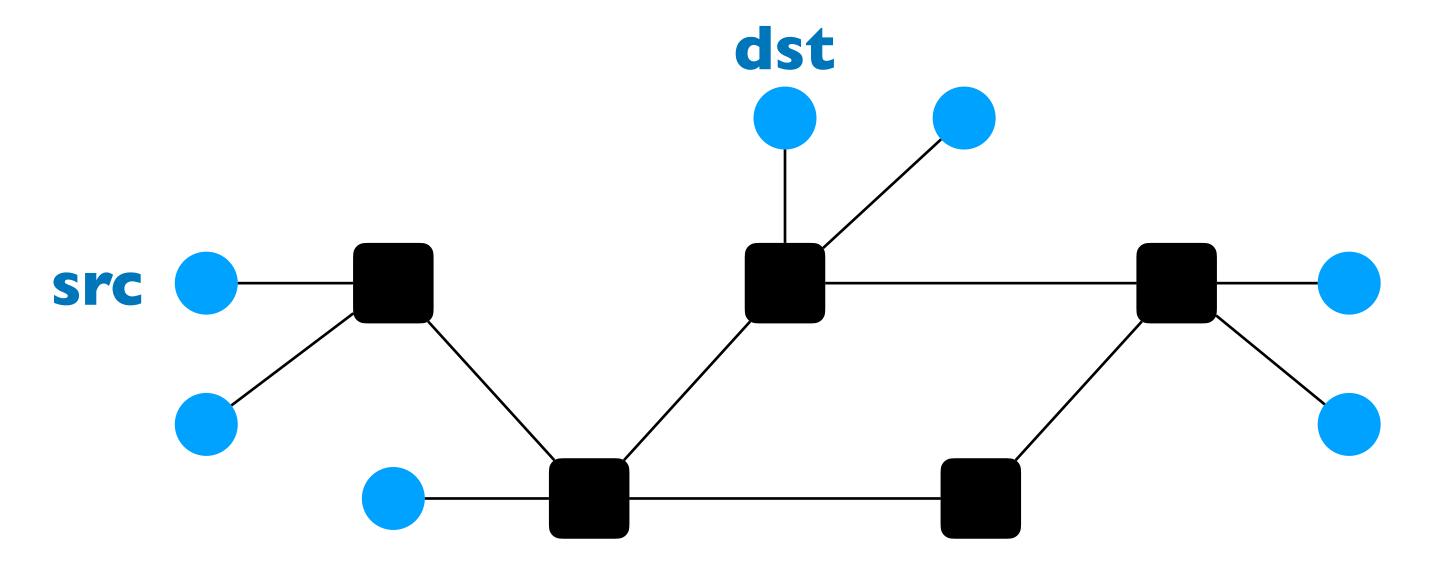
Two approaches to sharing

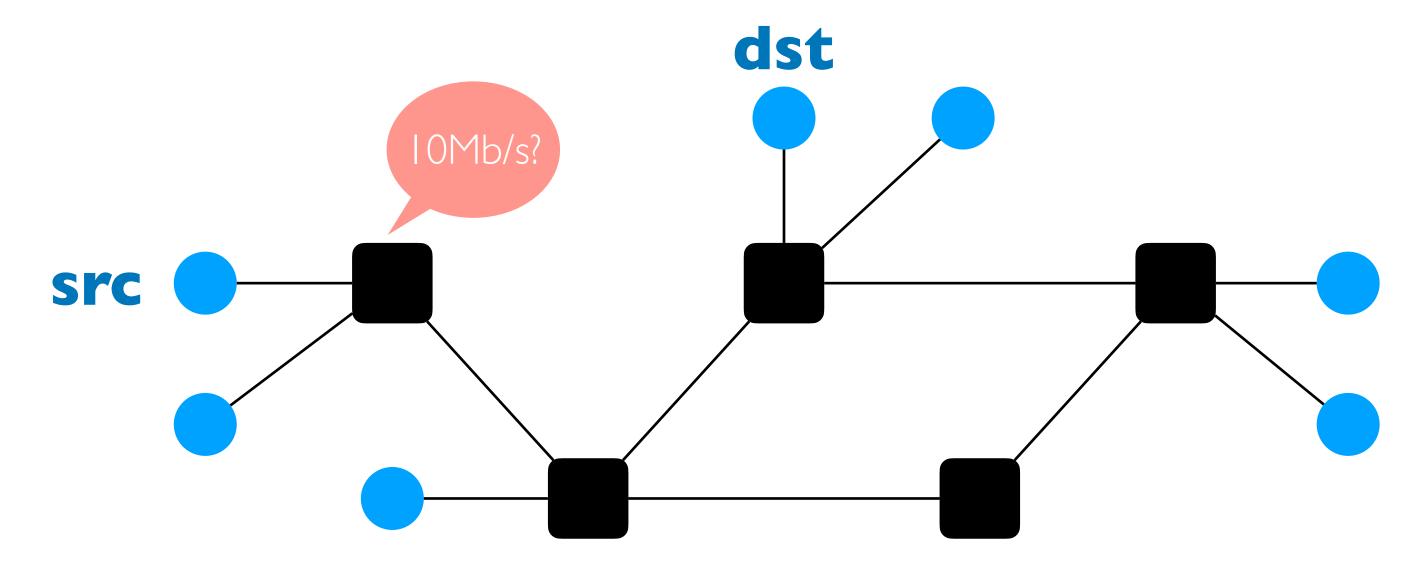
Packet switching

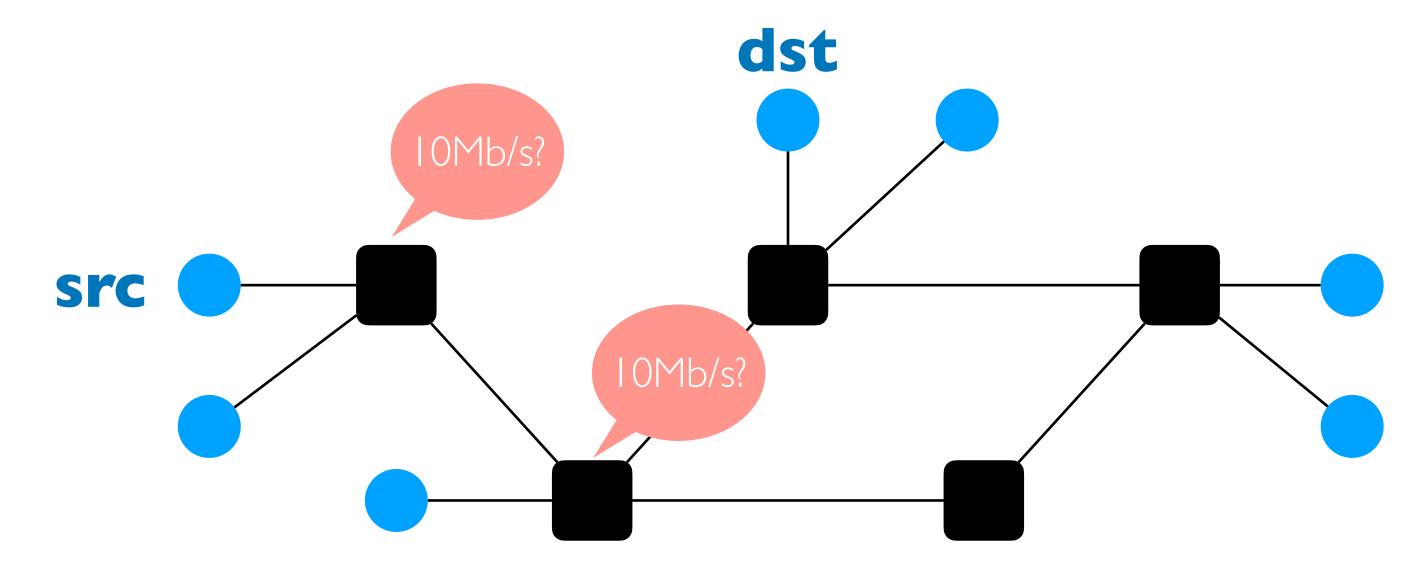
- Network resources consumed on demand per-packet
- "Admission control": per packet

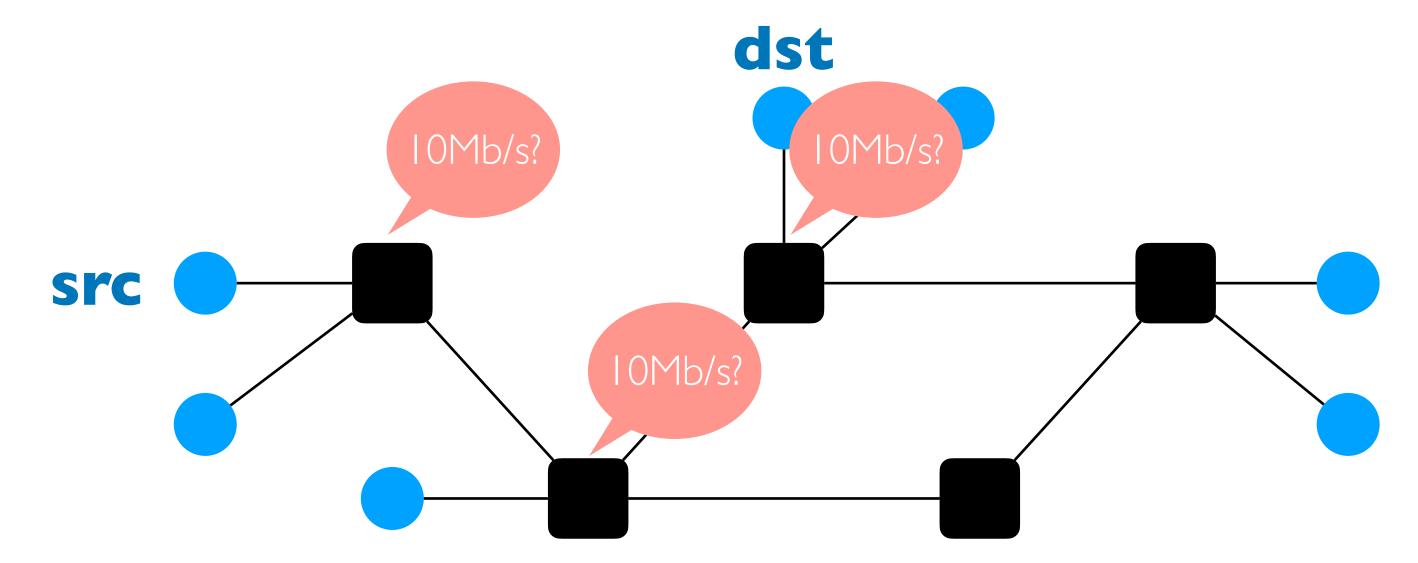
- Network resources reserved a priori at "connection" initiation
- "Admission control": per connection

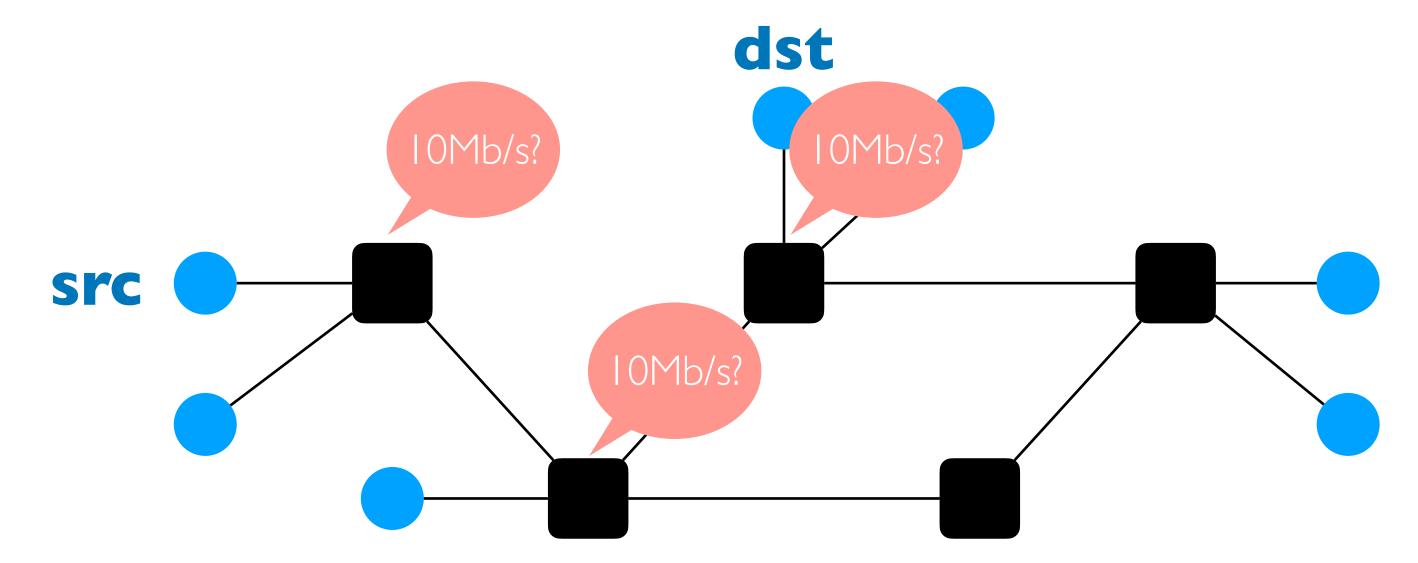




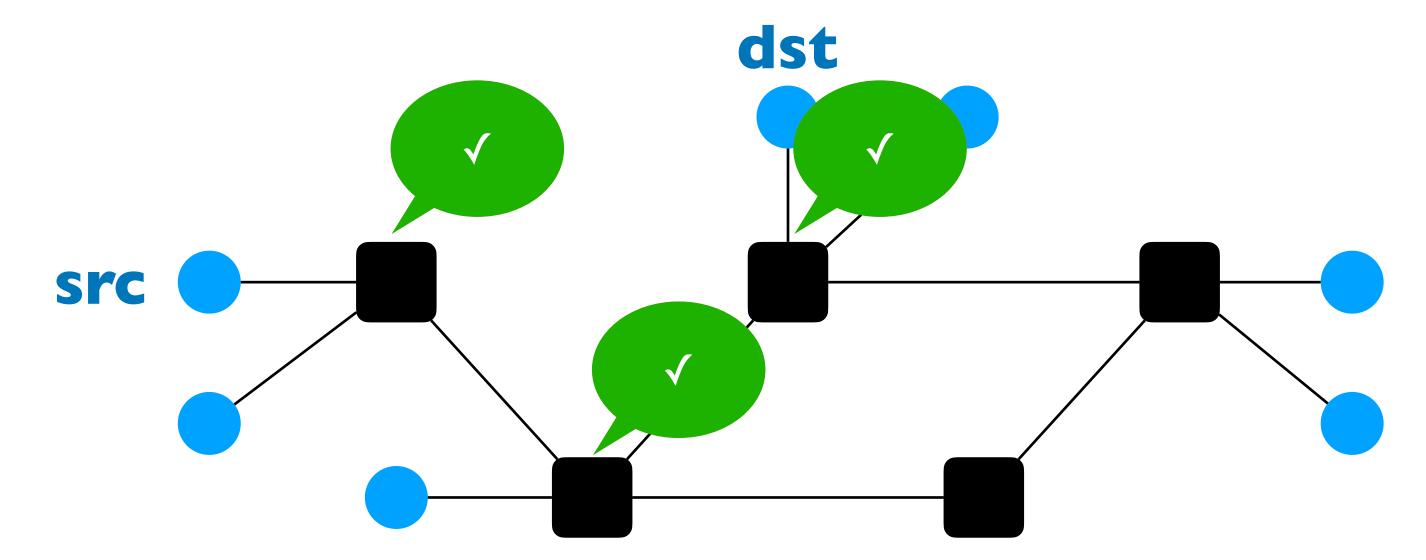




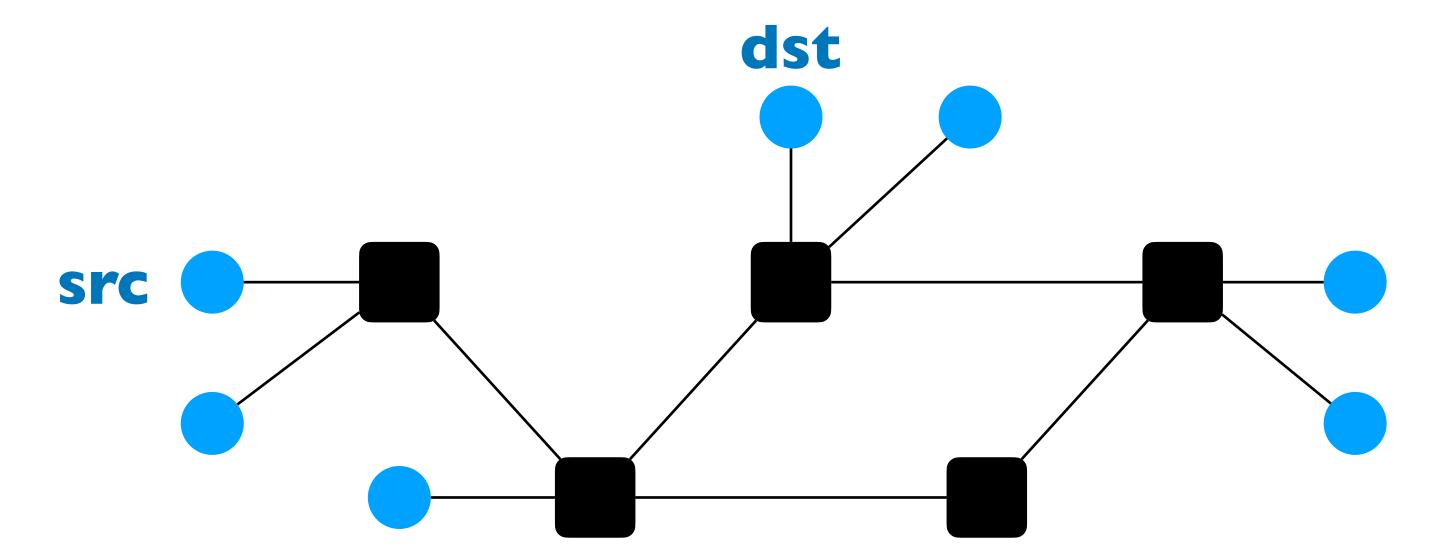




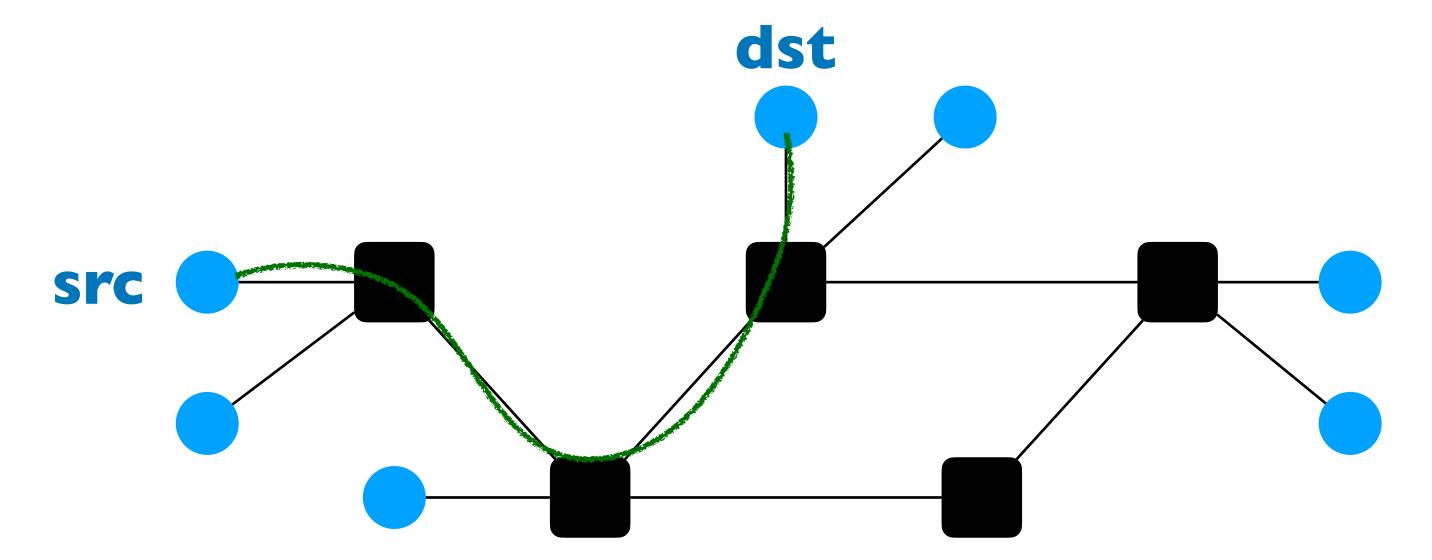
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- 2. Switches "establish a circuit"



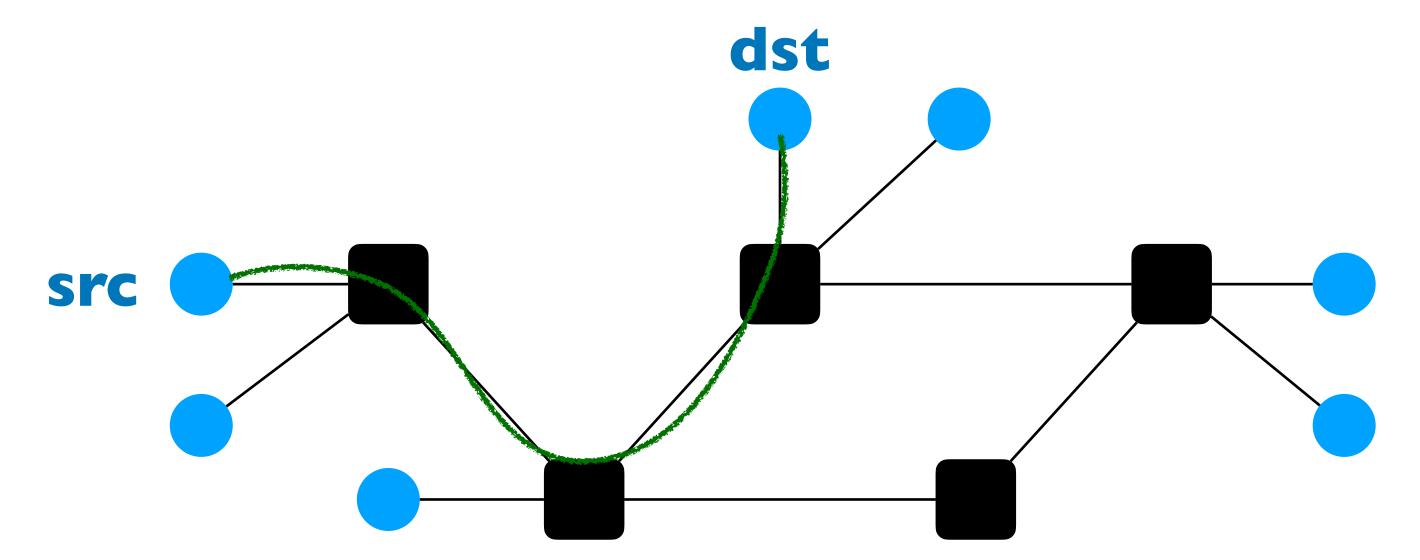
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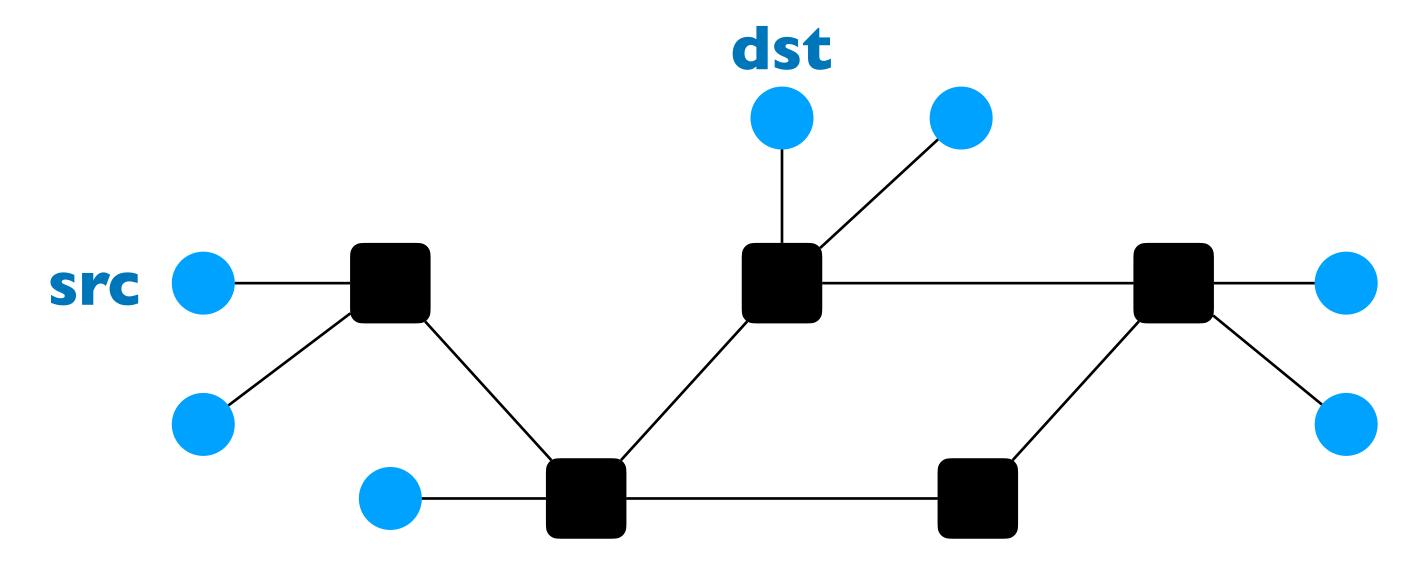
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- 1. The src sends a reservation request to dst
- 2. Switches "establish a circuit"
- 3. The **src** starts sending data
- 4. The src sends a "teardown circuit" message

Packet Switching

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- Each packet travels independently
- No link resources are reserved in advance
- Packet switching exploits statistical multiplexing better than circuit switching
 - Sharing using the statistics of demand
 - Good for bursty traffic (average << demand)

Circuit Switching

What's good?

- Predictable performance
- Simple/fast switching (once circuit is established)

What's not-so-good?

- Complexity of circuit setup/teardown
- Inefficient when traffic is bursty
- Circuit setup adds delay
- Switch fails -> its circuits fail

What's good?

- Efficient use of network resources
- Simpler to implement
- Robust: can "route around trouble"

What's not-so-good?

- Unpredictable performance
- Requires buffer management and congestion control

Performance Metrics

- **Delay:** How long does it take to send a packet from its source to destination?
- Loss: What fraction of the packets sent to the destination are dropped?
- Throughput: At what rate is the destination receiving data from the source?

A Network Link



Link bandwidth

• Number of bits sent/received per unit time

Propagation delay

• Time for one bit to move through the link (seconds)

Bandwidth-Delay Product (BDP)

- Number of bits "in flight" at any time
- BDP = Bandwidth x Propagation delay

Delay

- Consists of four components
 - Transmission Delay
 - Propagation Delay
 - Queueing Delay
 - Processing Delay

Delay

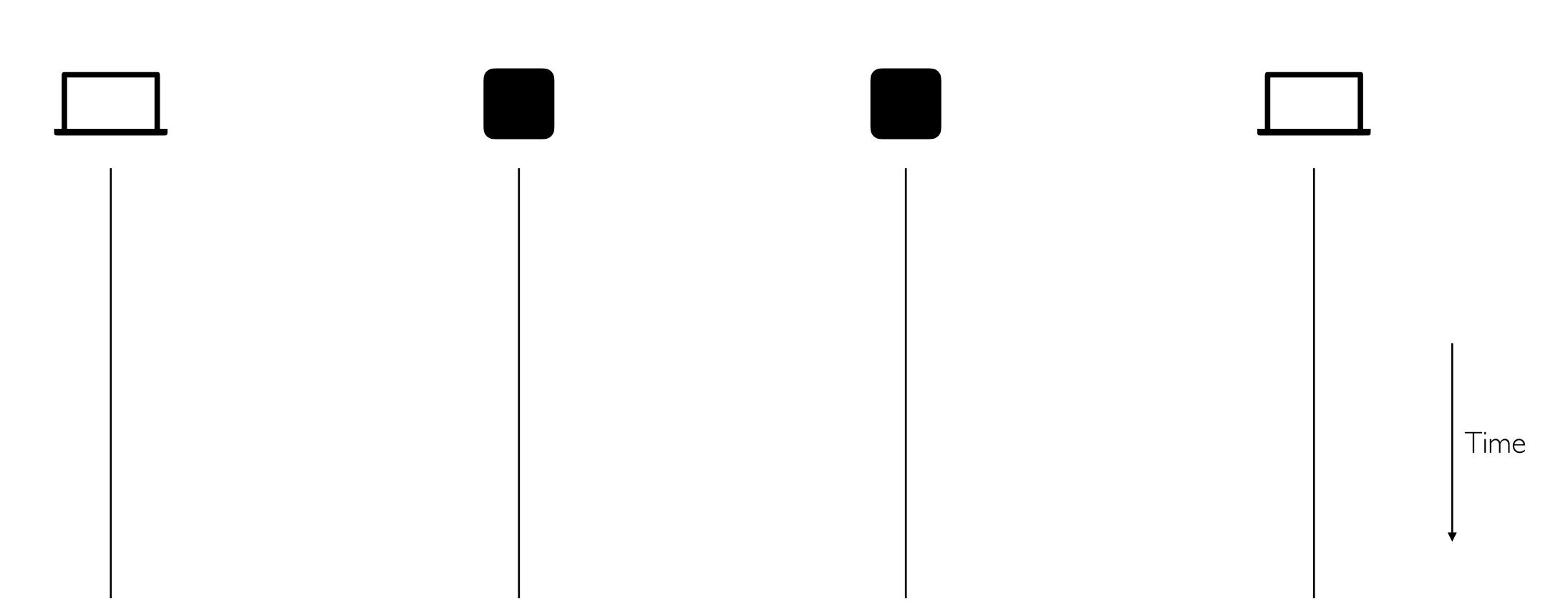
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Due to link properties

Due to traffic matrix and switch internals

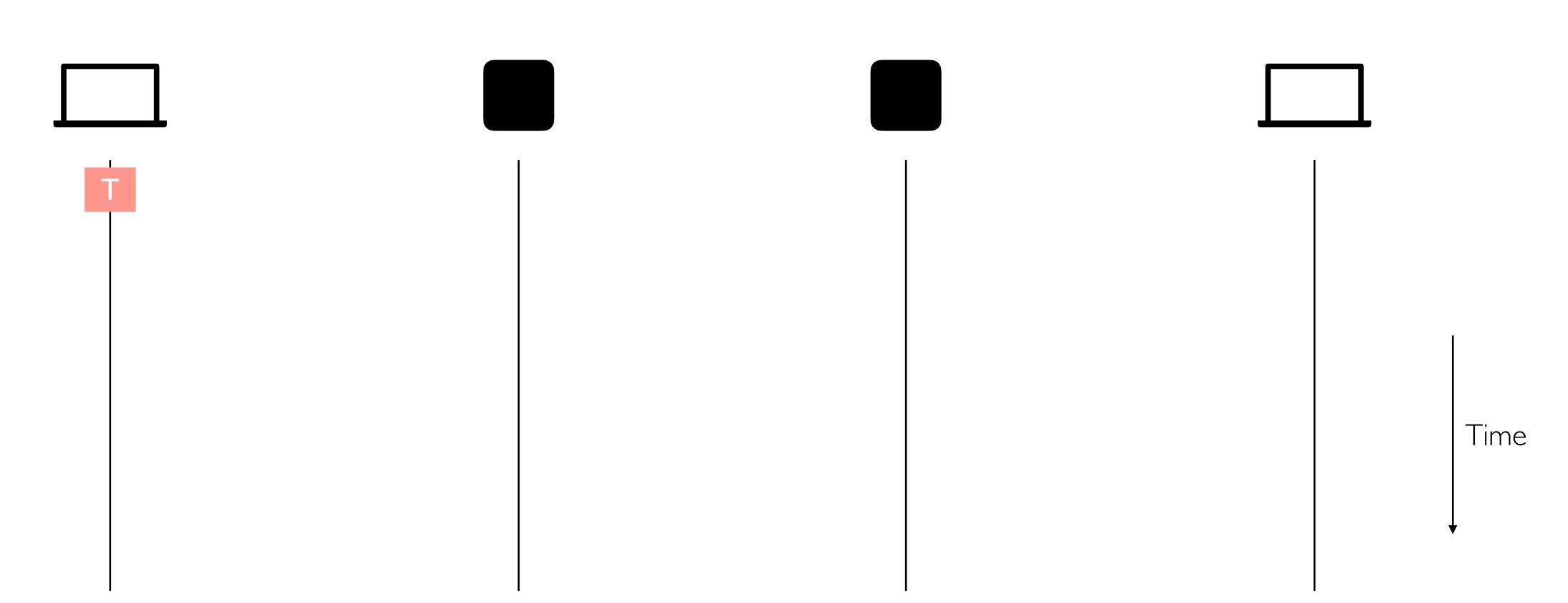
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 - Queuing Delay
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Pr Processing Delay



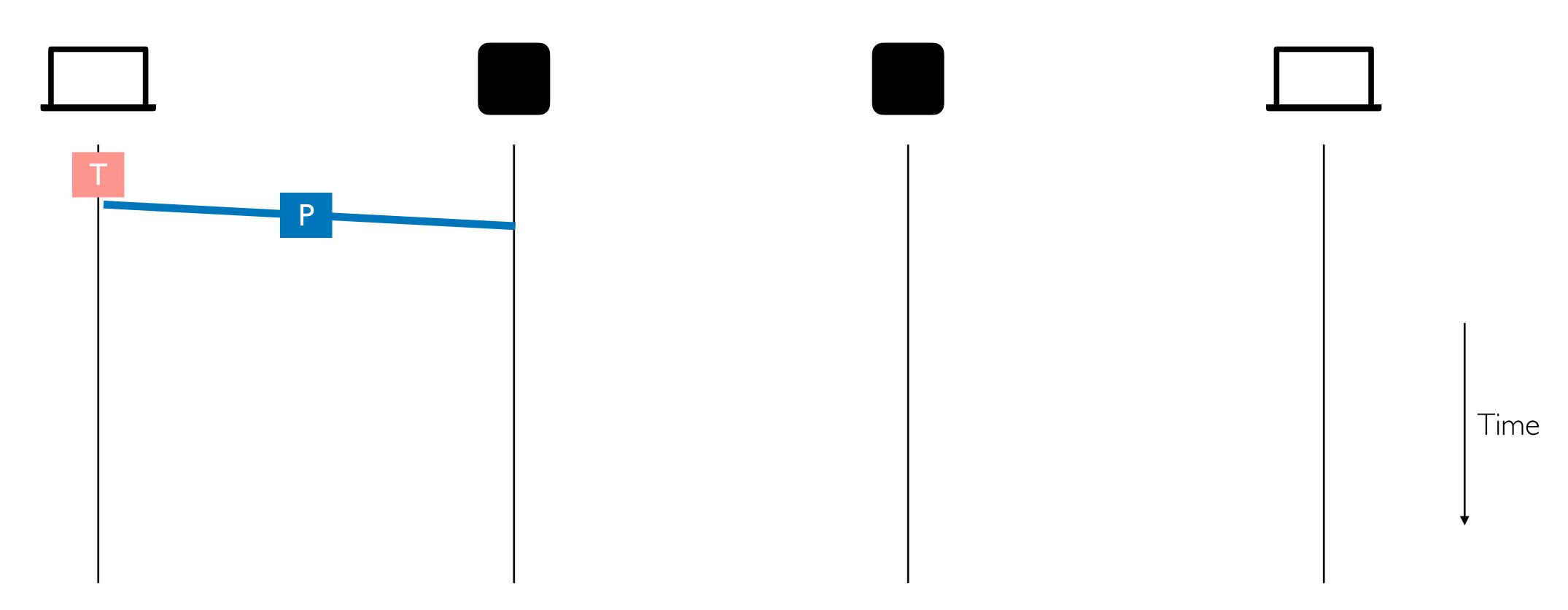
- T Transmission Delay
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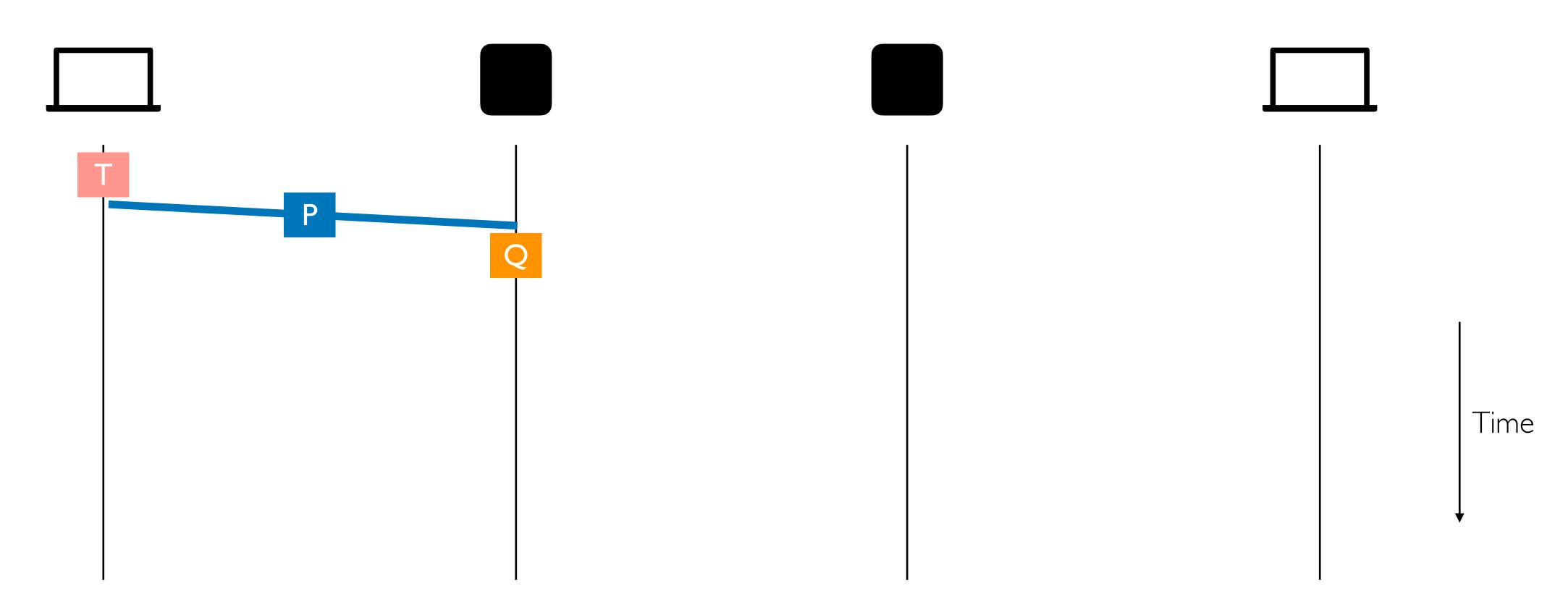
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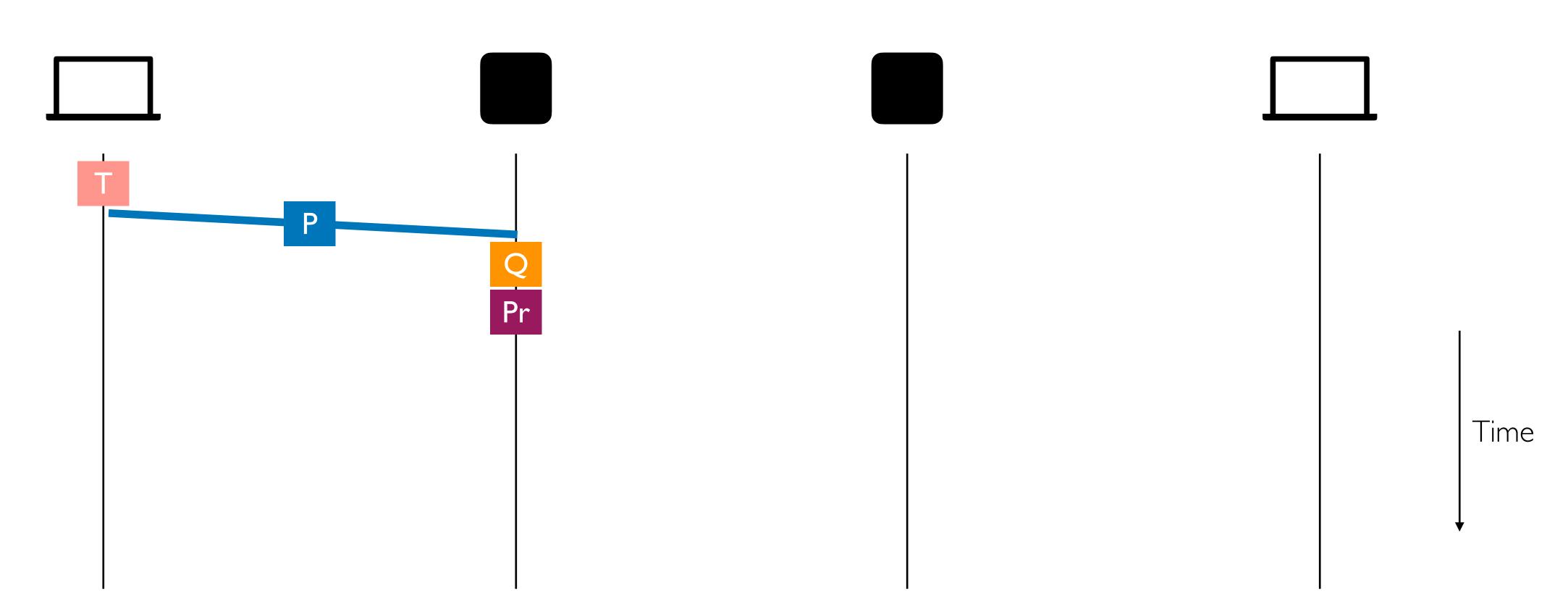
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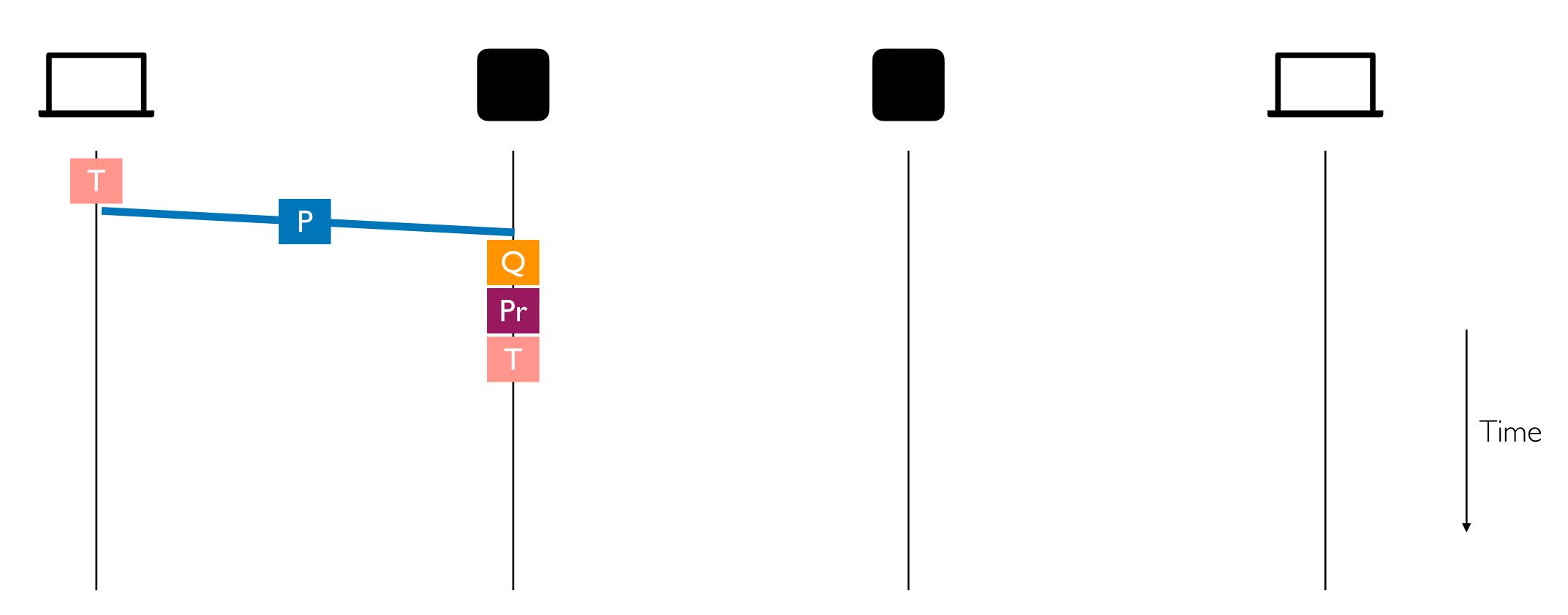
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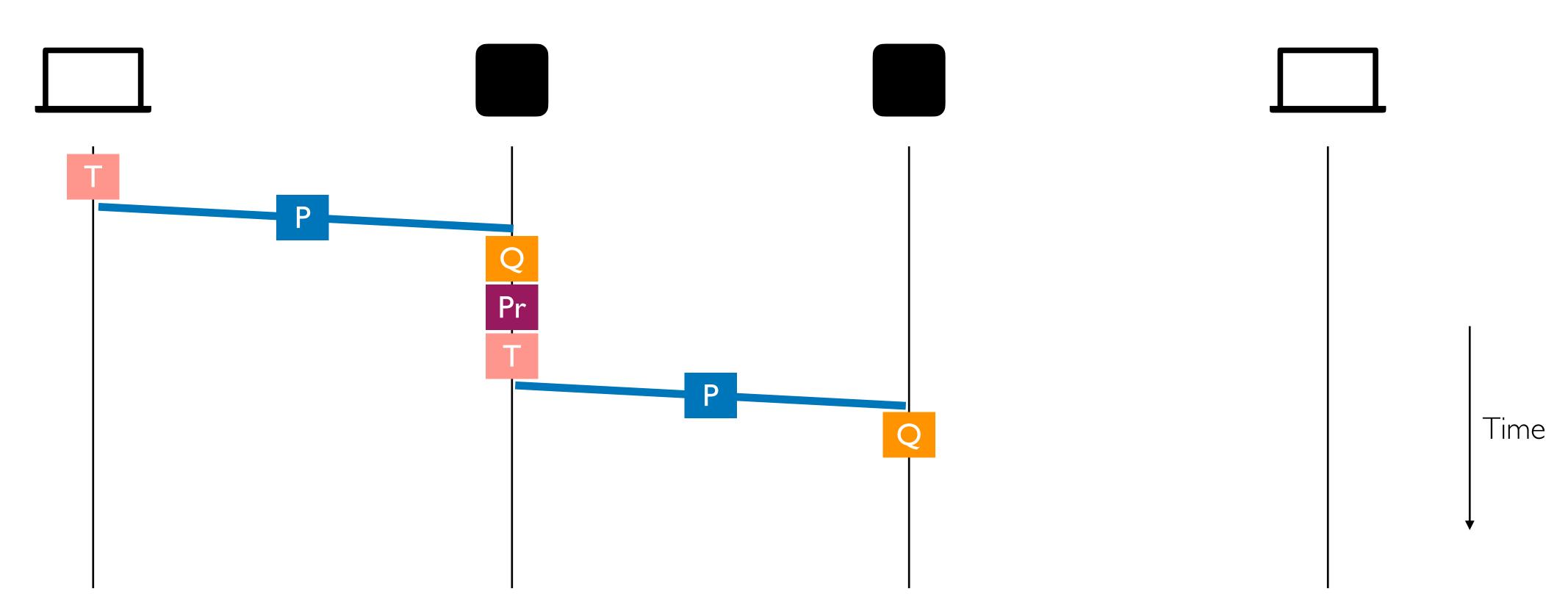
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Propagation Delay

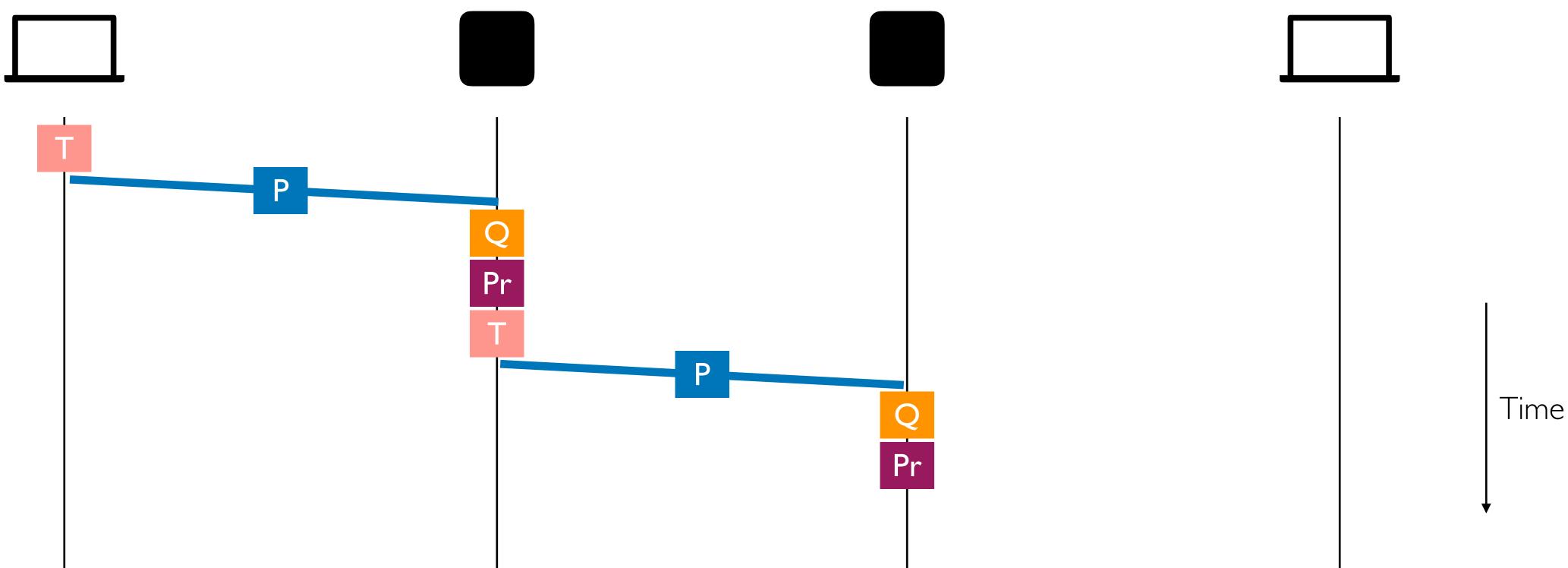
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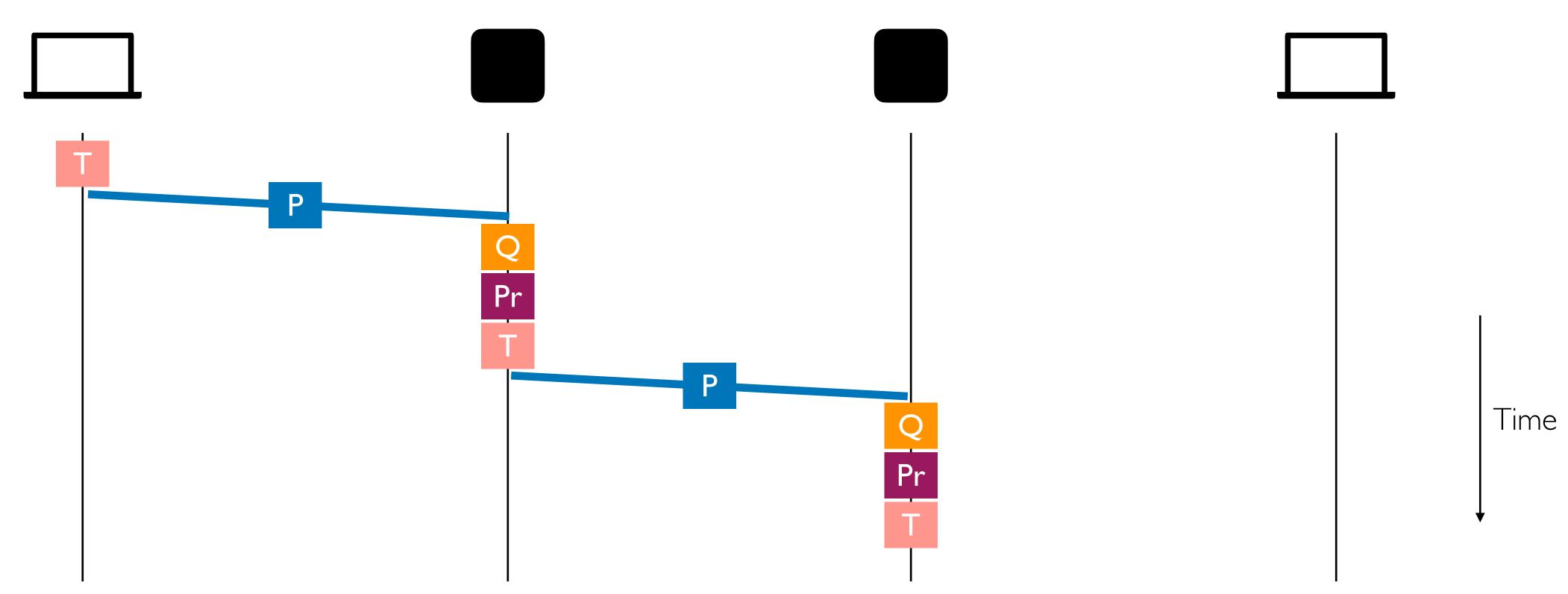
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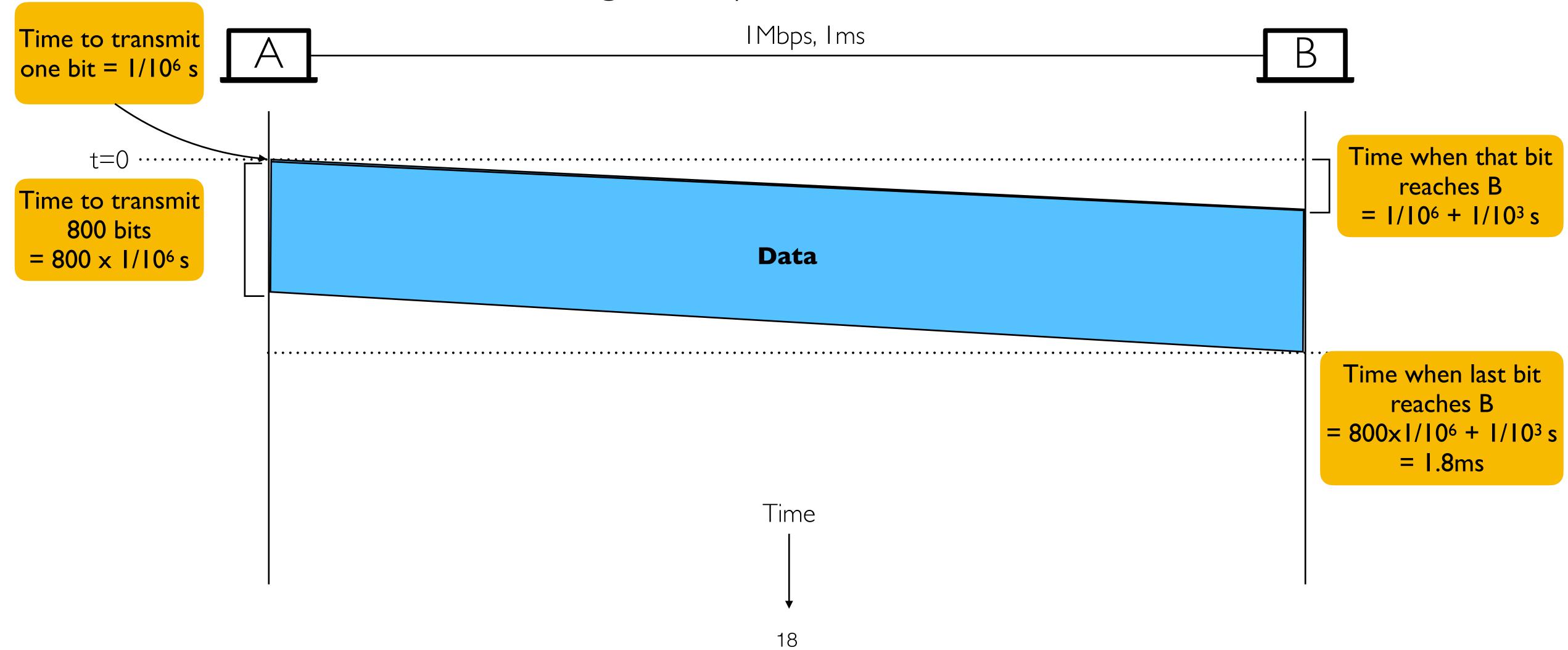
Processing Delay



Queuing Delay Transmission Delay Pr Processing Delay Propagation Delay Time

Packet Delay

Sending 100B packets from A to B?



Queueing Delay: Little's Law

A: avg. packet arrival rate (/s)

L: avg. # of packets waiting in queue

W: avg. time packets wait in queue

$$L = A \times W$$

Or,

$$W = L / A$$

Architecture

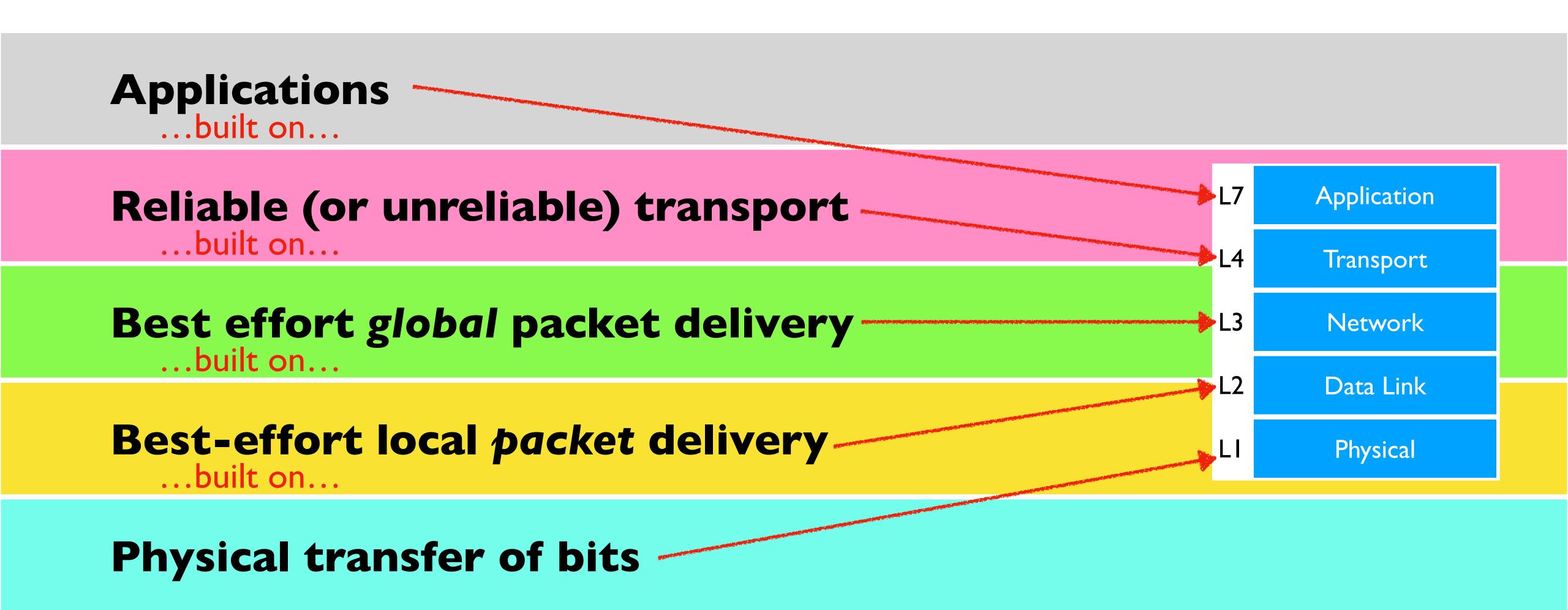
You should know:

- Layering: what/where/why
- Protocols: what/where/why
- Principles: layering, e2e argument, fate-sharing, "narrow waist"
- Benefits and weaknesses/consequences of principles/choices
 - E.g., layering is good because... but has hurt...

Layering

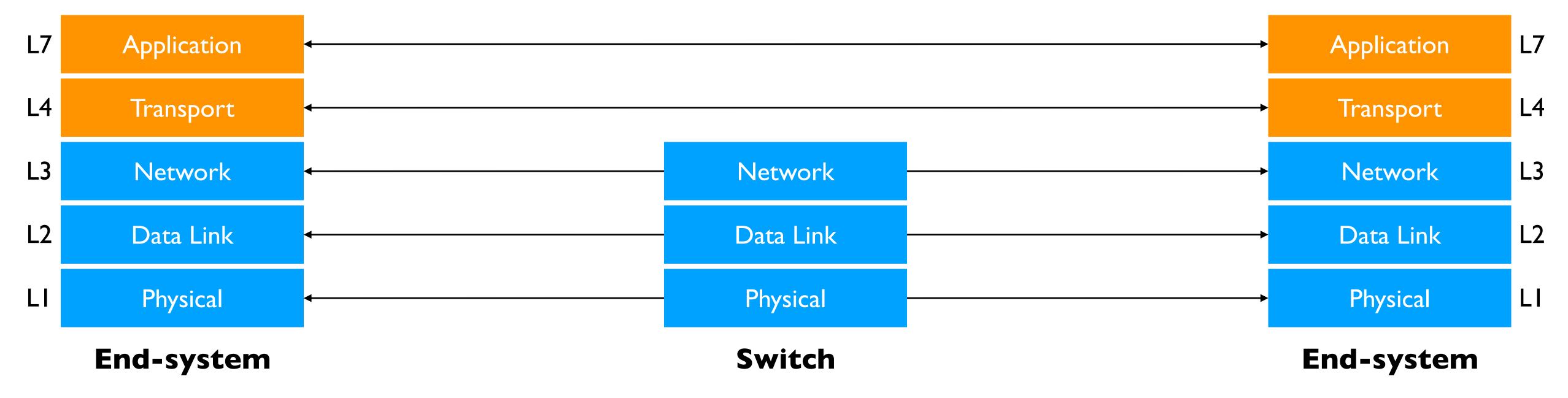
- Layering is a form of modularization
- System is broken into a vertical hierarchy of logically distinct entities (layers)
- Service provided by one layer is based solely on the service provided by layer below

Internet Layers



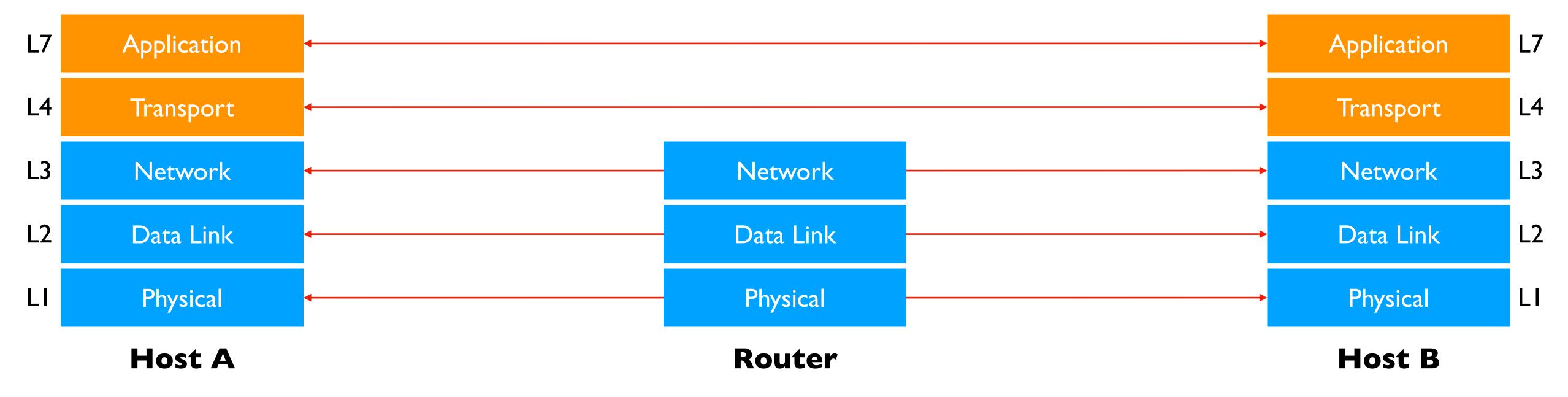
What gets implemented where?

- Lower three layers implemented everywhere
- Top two layers implemented only on hosts



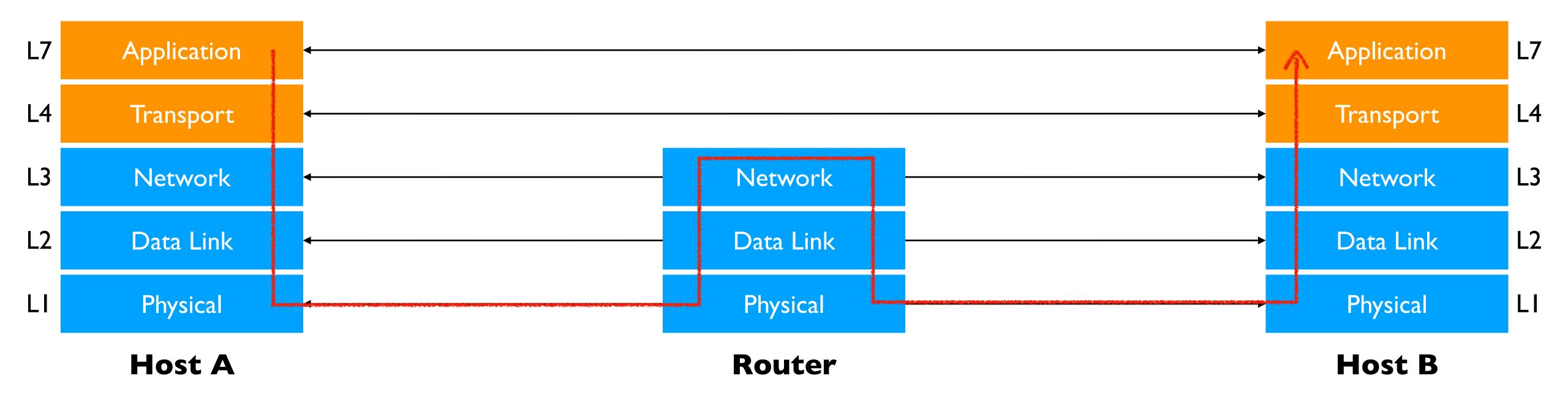
Logical Communication

Layers interact with peer's corresponding layer

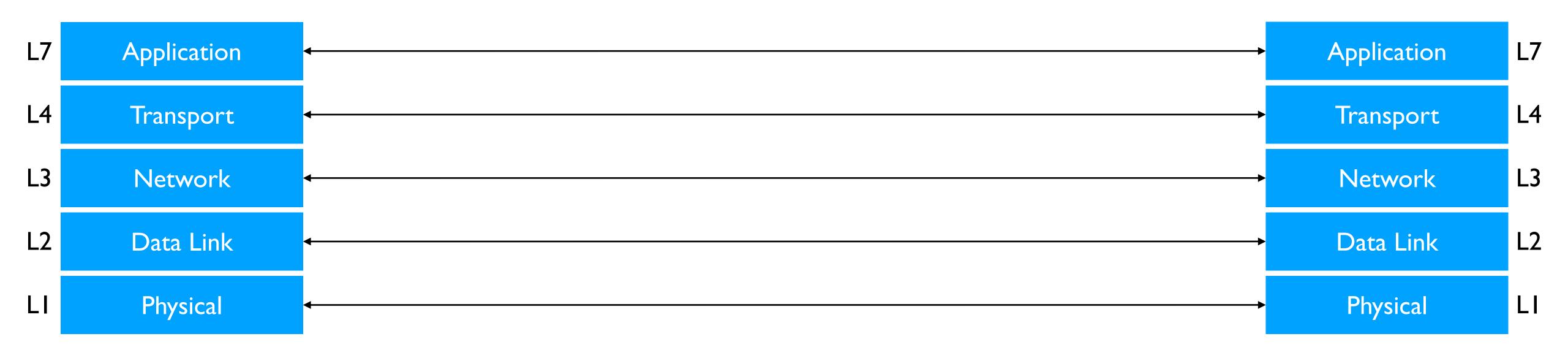


Physical Communication

- Communication goes down to the physical network
- Then up to relevant layer

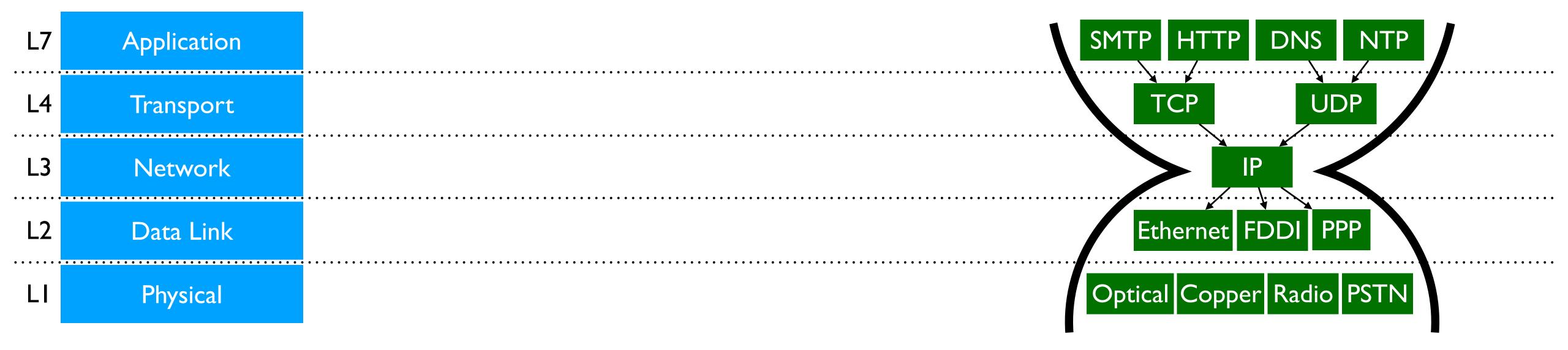


Layers and Protocols



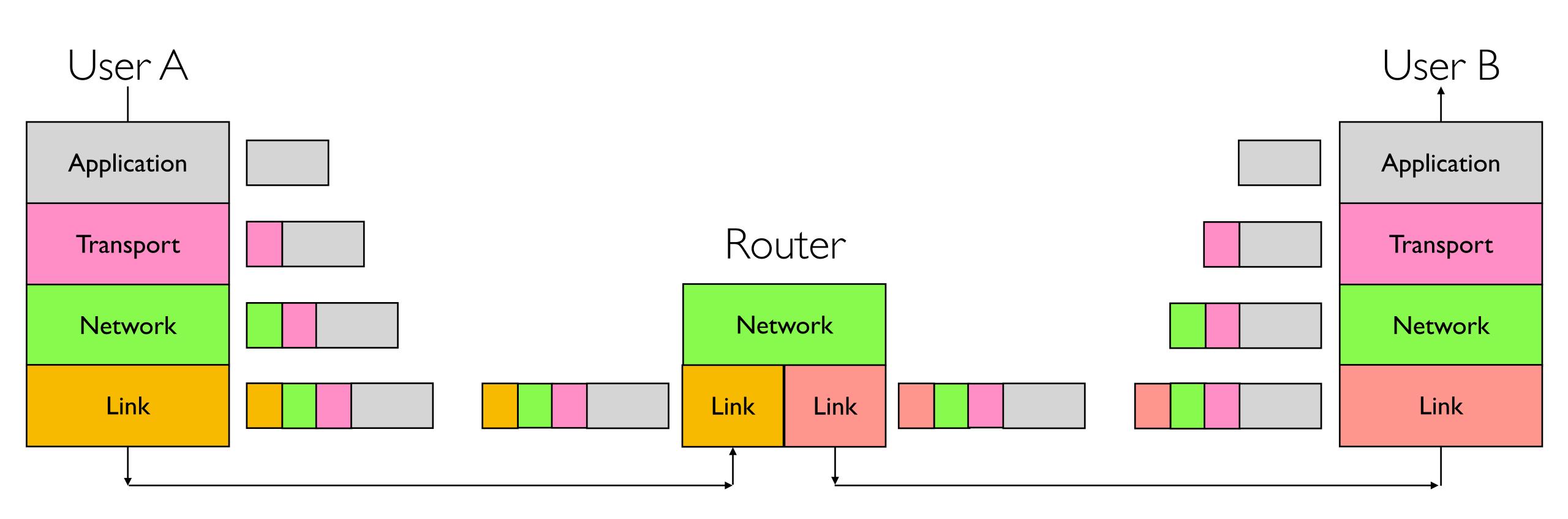
Communication between peer layers on different systems is defined by protocols

Protocols at different Layers



There's only one network layer protocol!

Layer Encapsulation



Layers: Pros & Cons

Why Layer?

- Reduce complexity
- Improve flexibility
 - Each layer can evolve independently
- Why not layer?
 - Sub-optimal performance
 - Cross-layer information often useful

End-to-end argument: Intuition

- Some application requirements can only be correctly implemented at the end-system
 - Reliability, security, etc.
- End-systems:
 - Can satisfy the requirement without network's help
 - Will/must do so, since they cannot rely on the network
- Put this functionality at end-system, unless...
 - Performance

Implications of the E2E argument

- In layered design, the E2E principle provides guidance on which layers are implemented where
- Key argument for why IP offers only "best effort" delivery (leading to "dumb network / smart ends")
 - Reliability implemented at the end-system (TCP)
 - Often credited as key to the Internet's success

Architectural Wisdom

- Layering
 - Reduce complexity, increase flexibility
- IP as the "narrow waist"
 - Eases interoperability
- "Smart ends, dumb network" (E2E argument)
 - No application knowledge in network → more general
 - Fate sharing: No/minimal state in network → more robust to failure

Network Layer

Forwarding vs. Routing

- Forwarding: "data plane"
 - Directing one data packet
 - Each router using local routing state
- Routing: "control plane"
 - Computing the forwarding tables that guide packets
 - Jointly computed by routers using a distributed algorithm

Routing: Basic concepts

- Valid routing state
- Convergence
- Least-cost paths

"Valid" Routing State

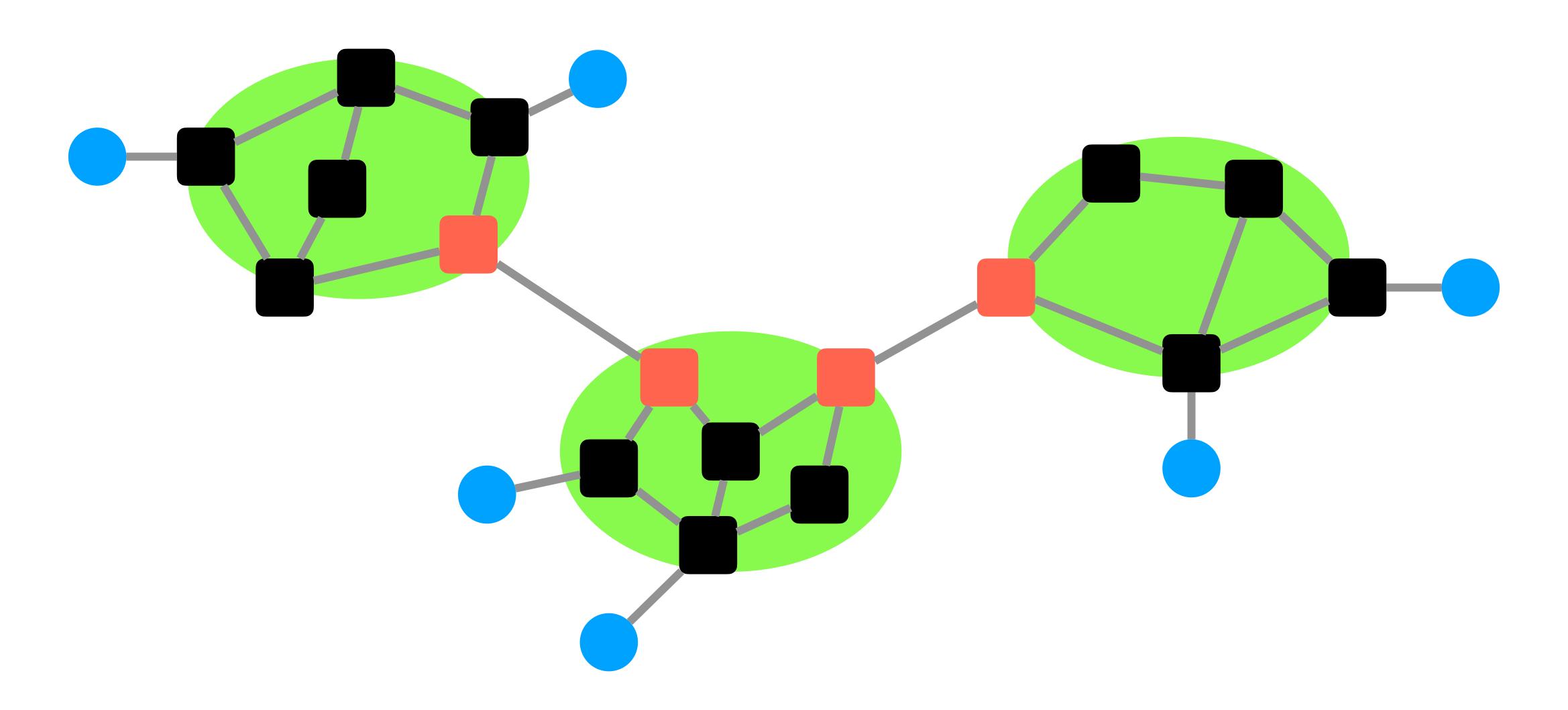
- Global state is "valid" if it produces forwarding decisions that always deliver packets to their destinations
- Global routing state is "valid" if and only if:
 - There are no dead-ends (other than the destination)
 - There are no loops

Convergence Delay

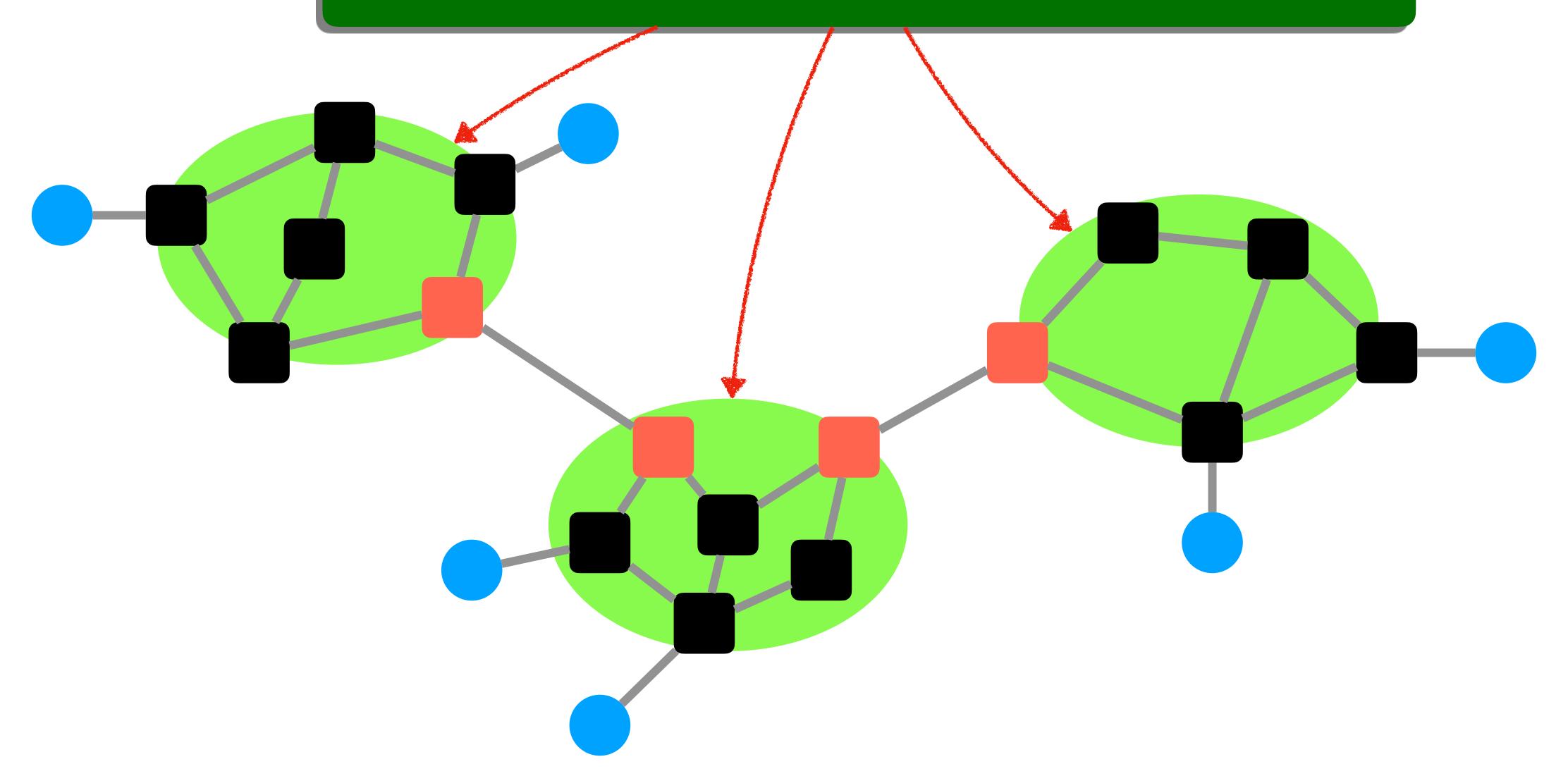
- Time to achieve convergence
 - E.g., all nodes have the same link-state database
- Sources of convergence delay?
 - Time to detect failure
 - Time to flood link-state information
 - Time to recompute forwarding tables
- Performance during convergence period
 - Lost packets due to black-holes
 - Looping packets
 - Out-of-order packets reaching the destination

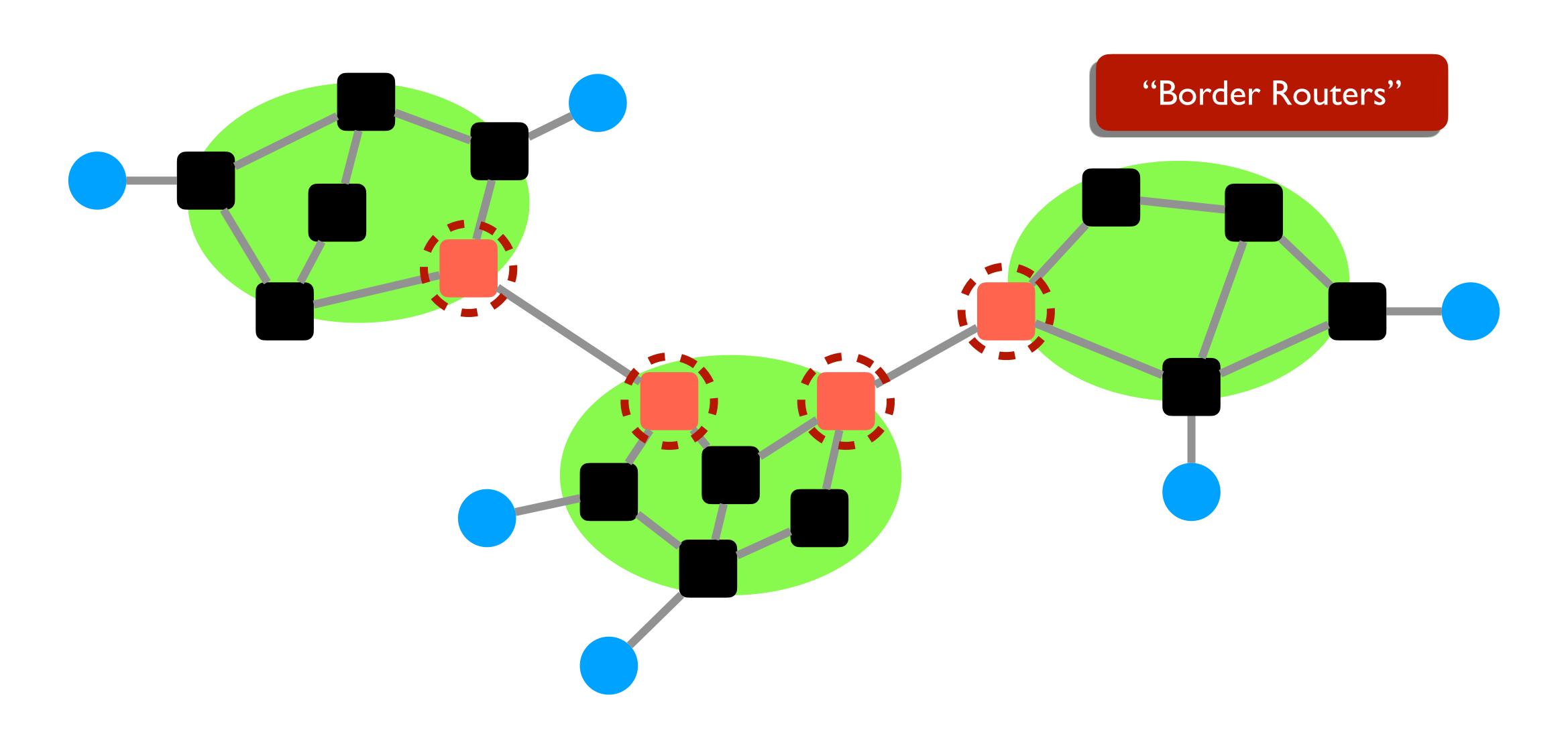
Least-cost path routing

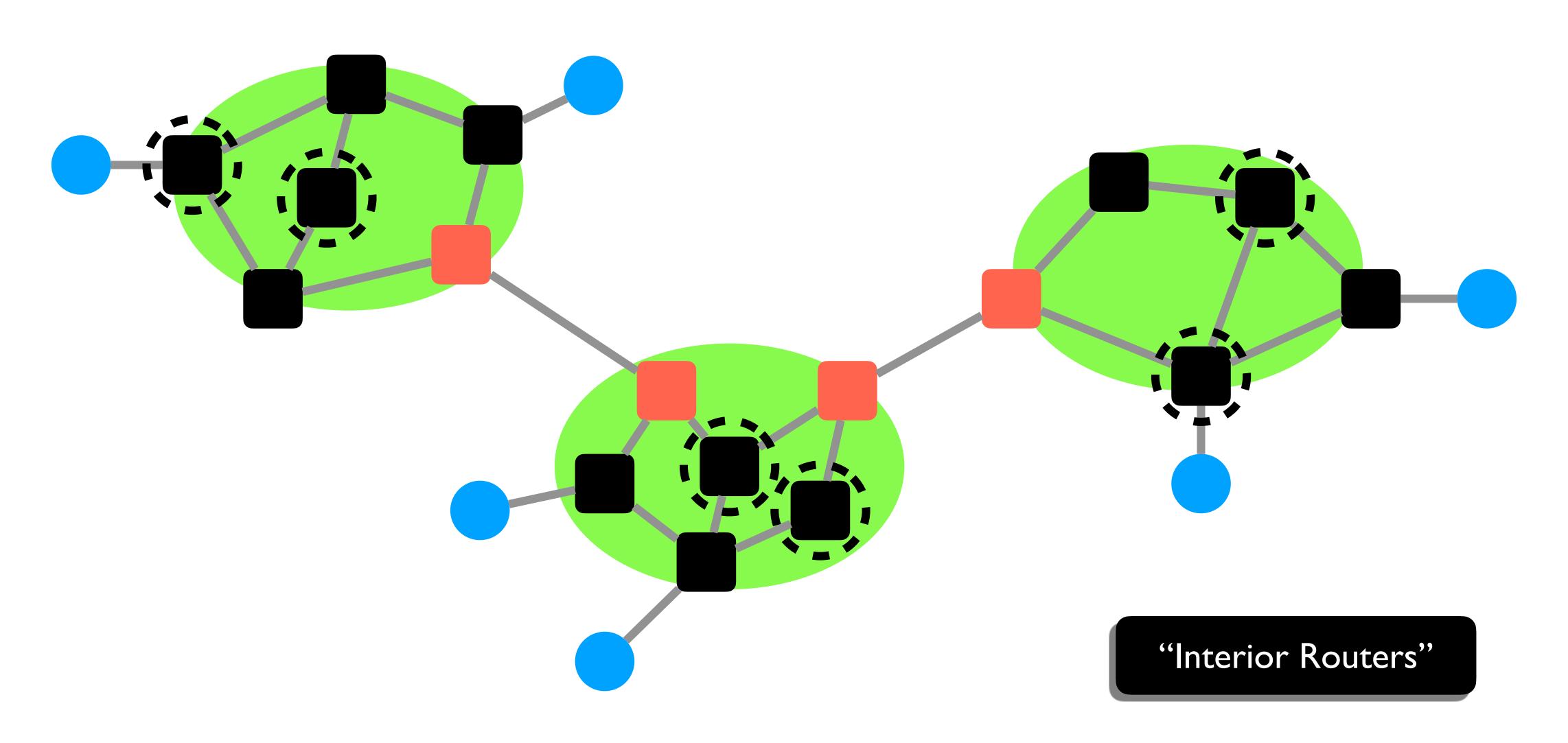
- Given: Router graph & link costs
- **Goal:** find least-cost path from each source router to each destination router
- "Least cost" routes are an easy way to avoid loops
 - No sensible cost metric is minimized by traversing a loop
- Least cost routes are also "destination-based"
 - i.e., they do not depend on the source
- Least cost paths form a spanning tree

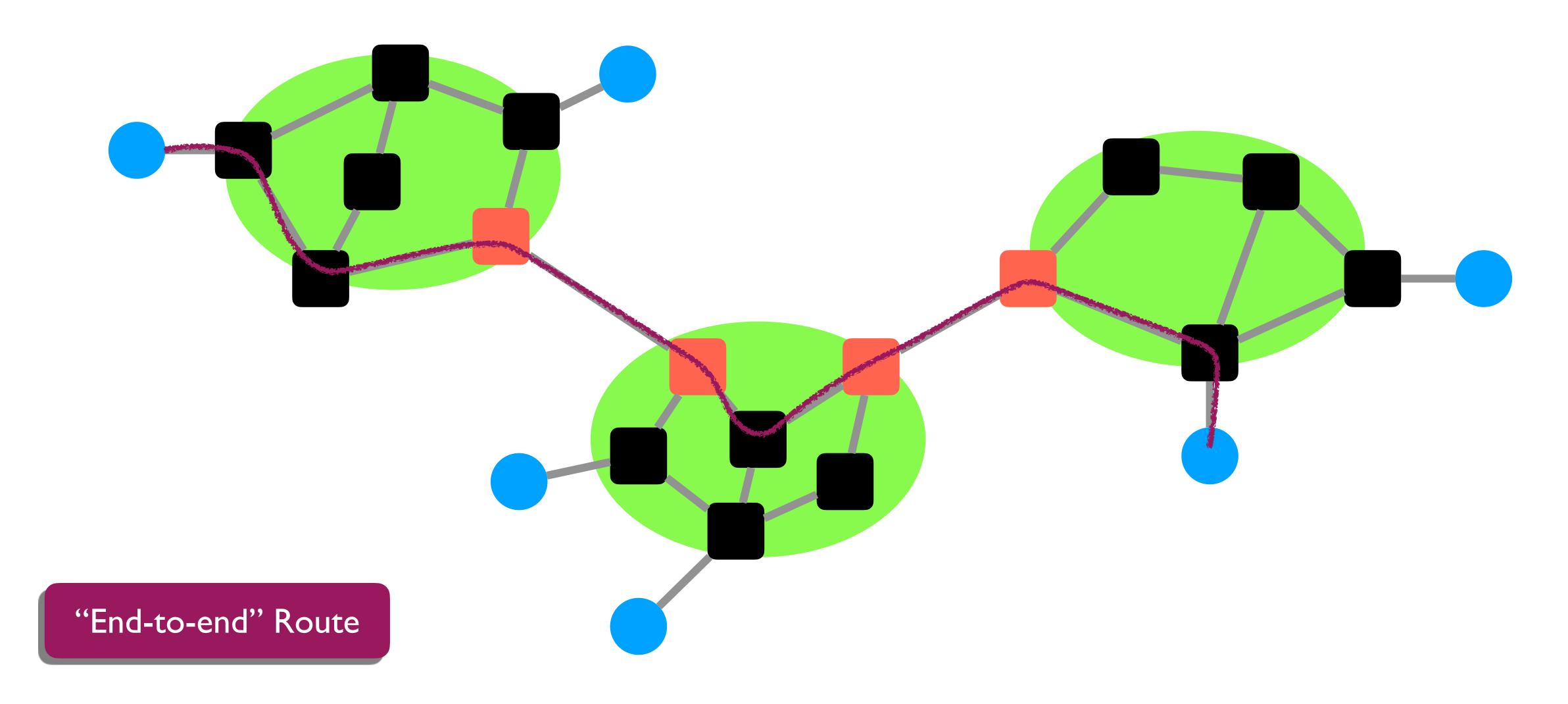


"Autonomous System (AS)" or "Domain" Region of a network under a single administrative authority









Internet Routing

- Internet routing works at two levels
- Each AS runs an **intra-domain** routing protocol that establishes routes within its domain
 - Intra-domain routes are "least cost"
 - E.g., Link State (OSPF) and Distance Vector (RIP)
- ASes participate in an **inter-domain** routing protocol that establishes routes between domains
 - Inter-domain routes determined by policy (need not be least-cost)
 - E.g., Path Vector (BGP)

• Every router knows its local "link state"

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- A router floods its link state to all other routers

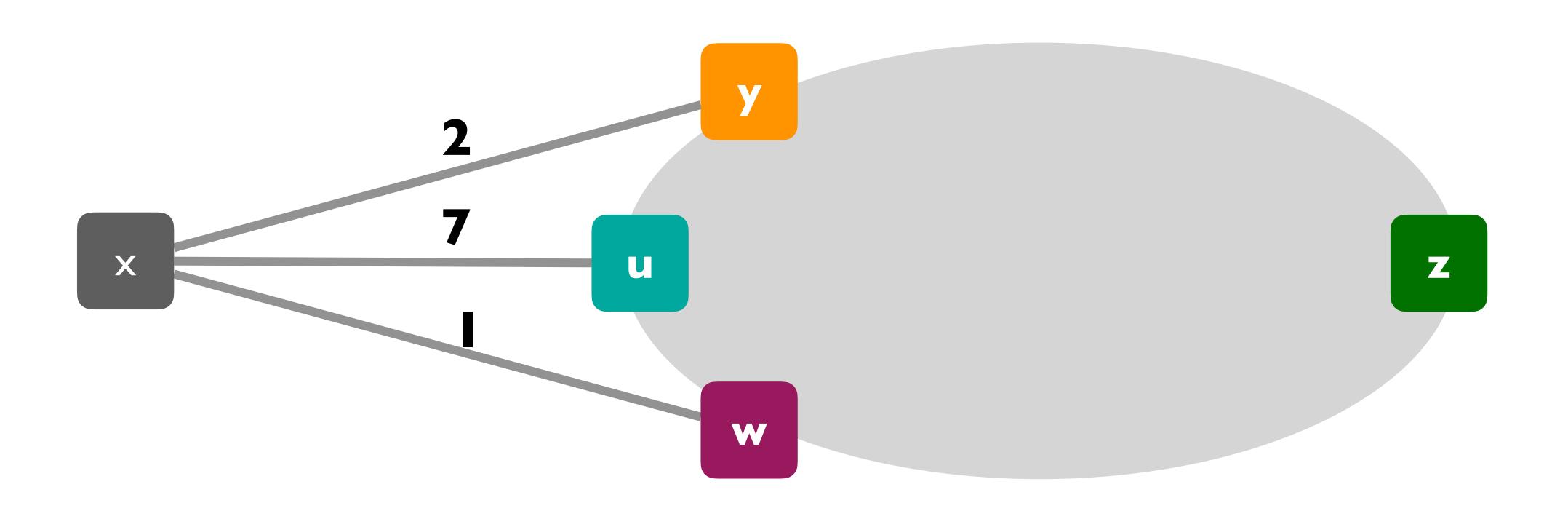
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- Hence, every router learns the entire network graph
- Every router locally runs Dijkstra's to compute its forwarding table

Distributed algorithm

- Distributed algorithm
- All routers run it "together"
 - Each router runs its own instance
 - Neighbors exchange and react to each other's messages

- Each router knows the links to its neighbors
- Each router has provisional "least cost" estimate to every other router its distance vector (DV)
 - E.g., Router A: "A can get to B with cost 11"
- Routers exchange this DV with their neighbors
- Routers look over the set of options offered by their neighbors and select the best one
- Iterative process converges to set of shortest paths



$$d_{x}(z) = \min_{n} \{ cost(x, n) + d_{n}(z) \}$$
For all neighbors n

Bellman-Ford Equation

DV Routing: Problems & Solutions

Problem: Count-to-infinity

- Cause:
 - z routes through y, y routes through x (to reach dst)
 - y loses connectivity to x
 - y decides to route through z (to reach dst)
- Can take a very long time to resolve

Solution: Split Horizon with Poisoned Reverse

- How:
 - If z routes to dst through y, z advertises to y that its cost to dst is infinite
 - y never dictates to route to dst through z
- Often avoids the count-to-infinity problem

Addressing Goal: Scalable Routing

- State: Small forwarding tables at routers
 - Much less than the number of hosts
- Churn: Limited rate of change in routing tables

Ability to aggregate addresses is crucial for both (One entry to summarize many addresses)

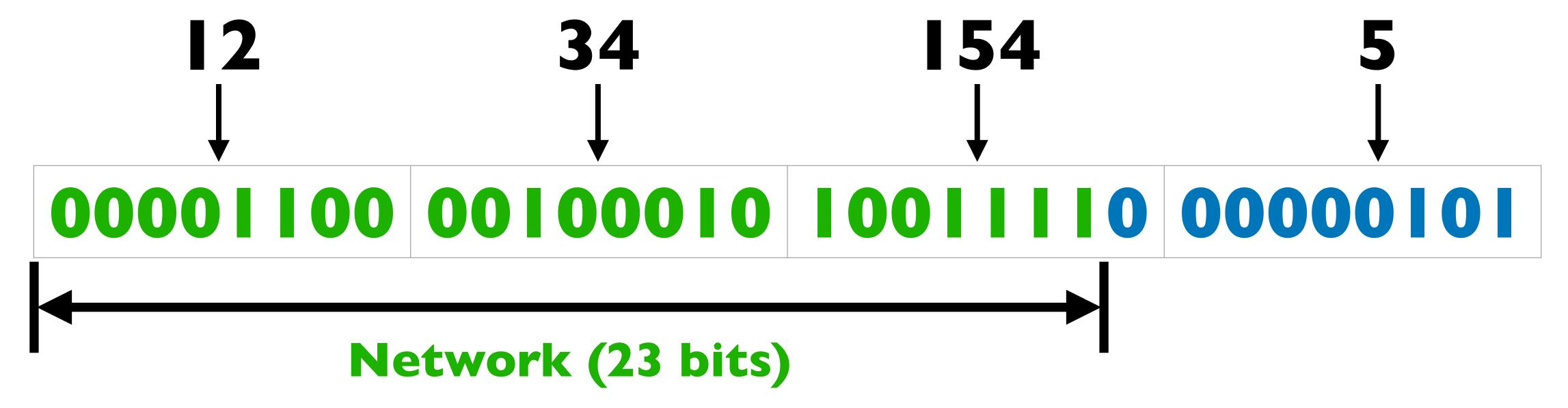
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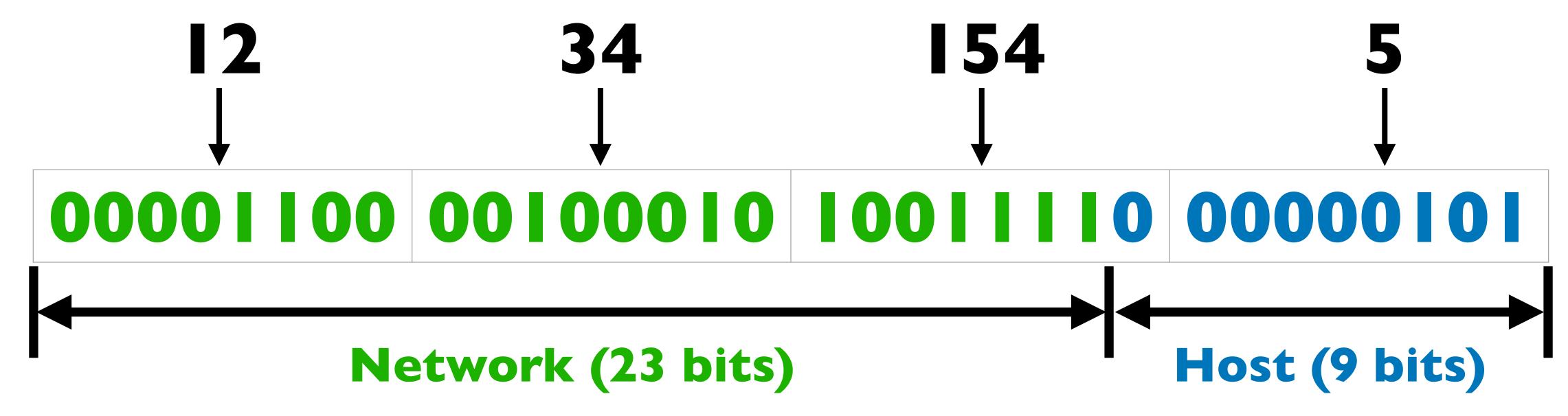
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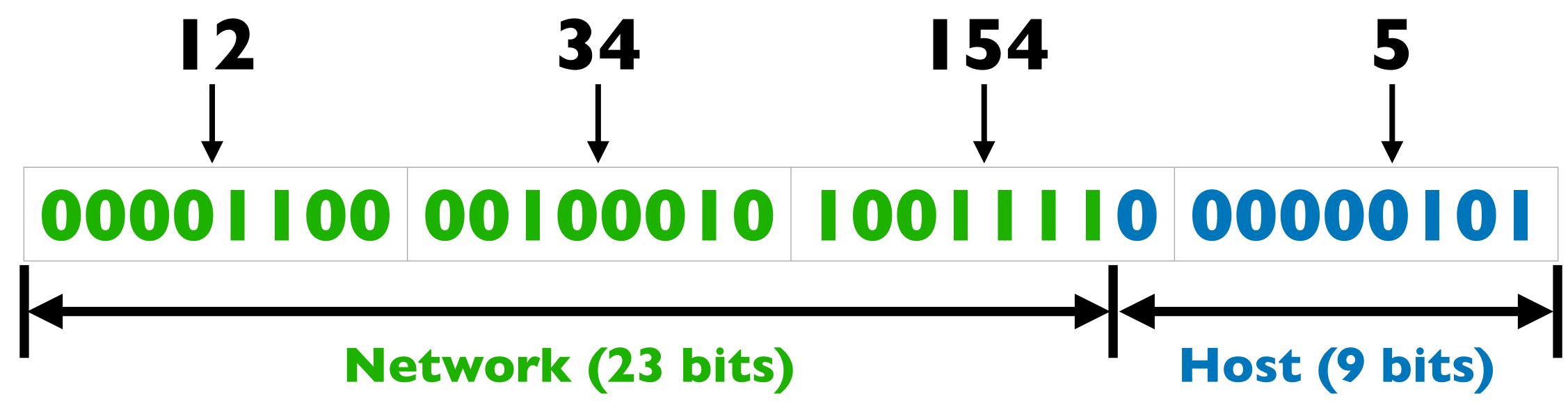
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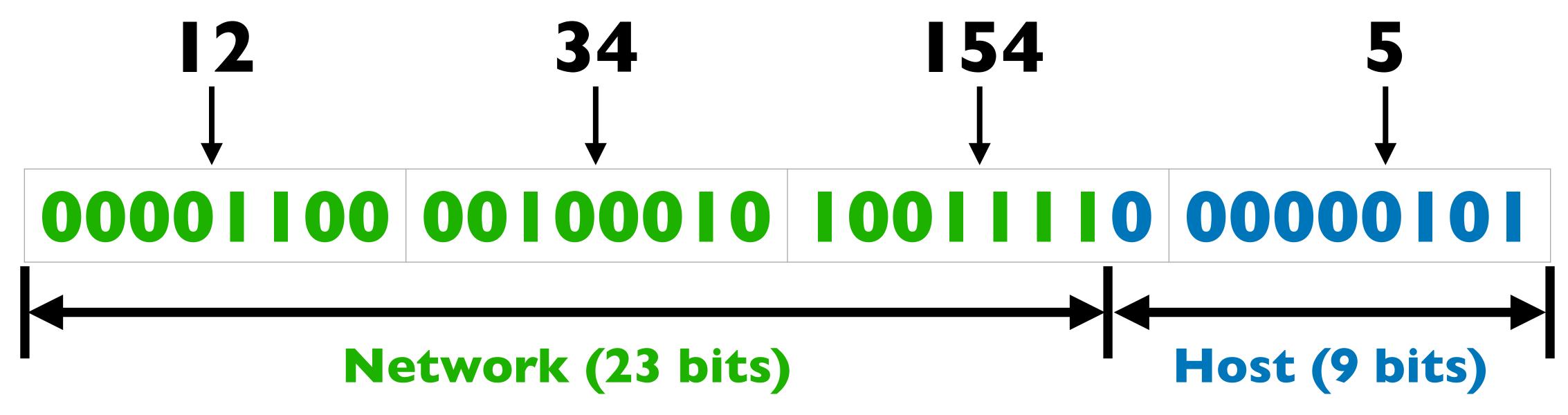


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• Interdomain routing operates on the network prefix

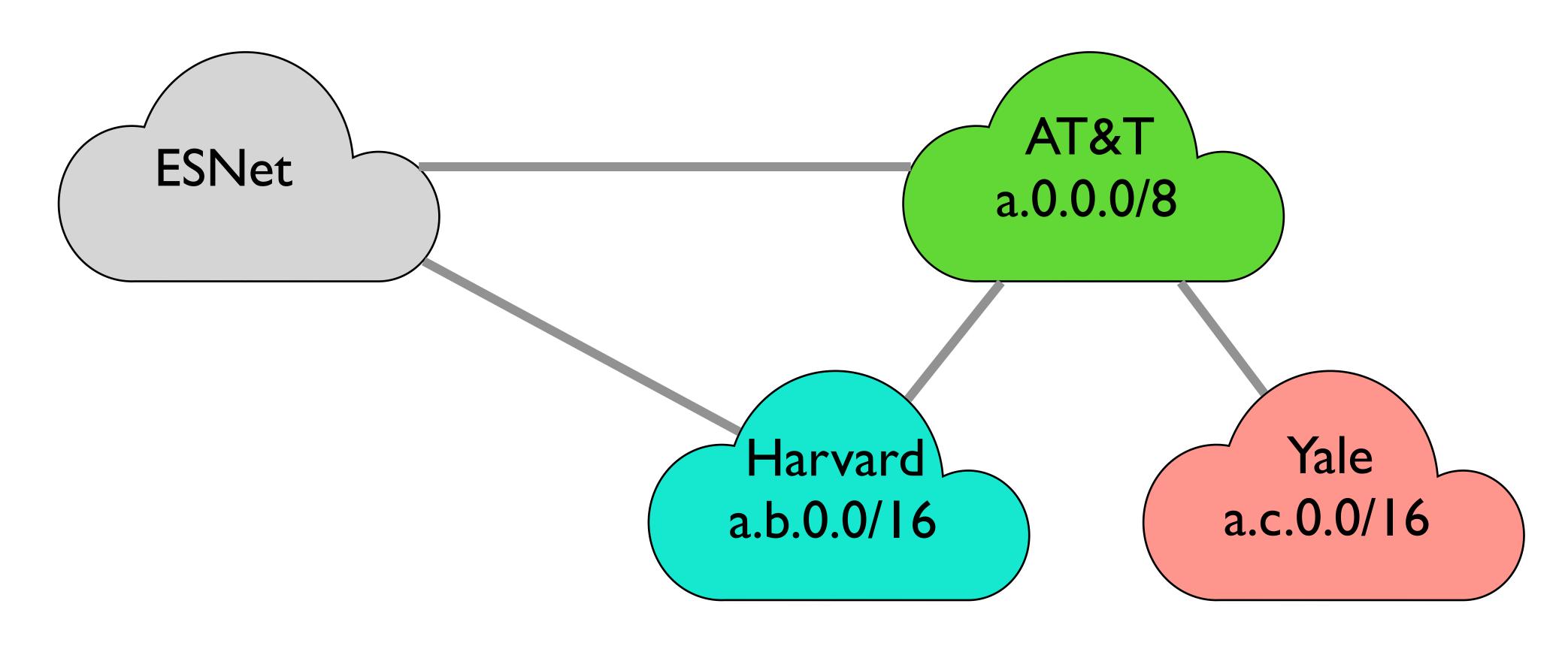
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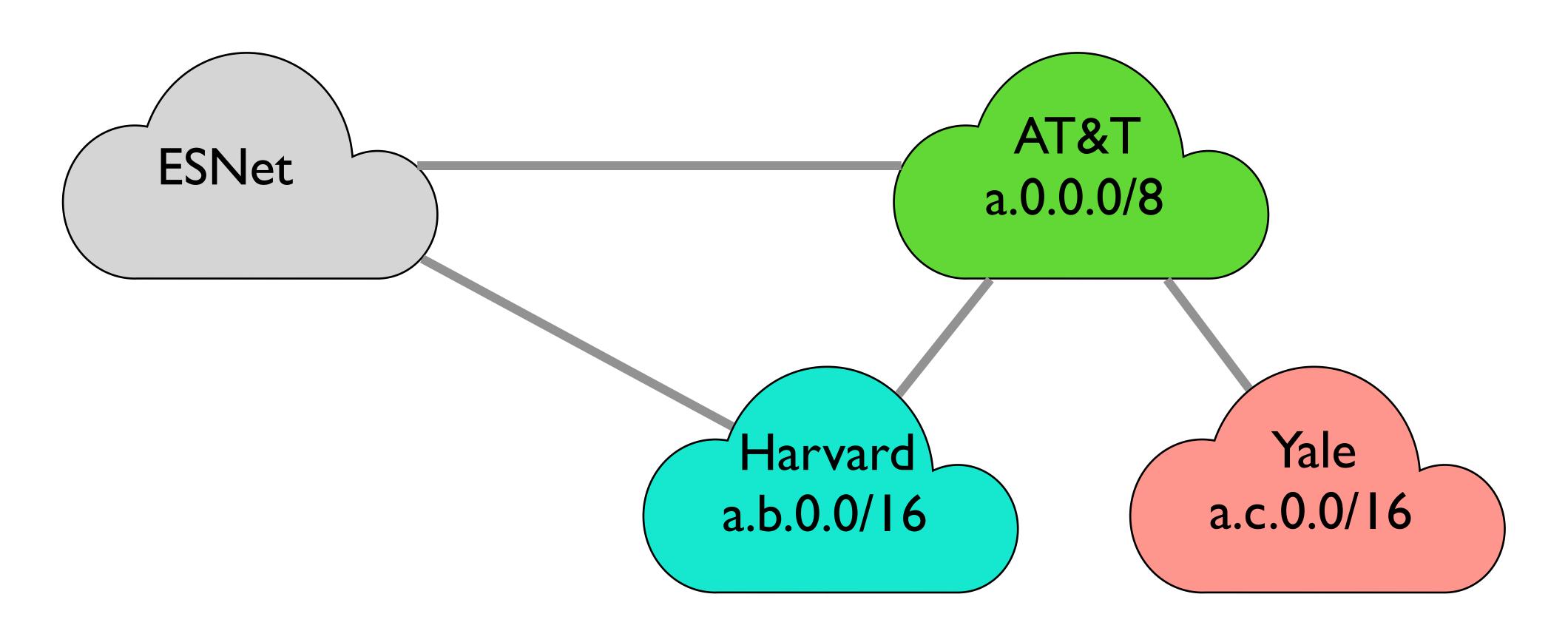
- Interdomain routing operates on the network prefix
- "Slash" notation: 12.34.158.0/23 \rightarrow network with a 23 bit prefixes and 29 host addresses

Hierarchical address allocation only helps routing scalability if allocation matches topological hierarchy

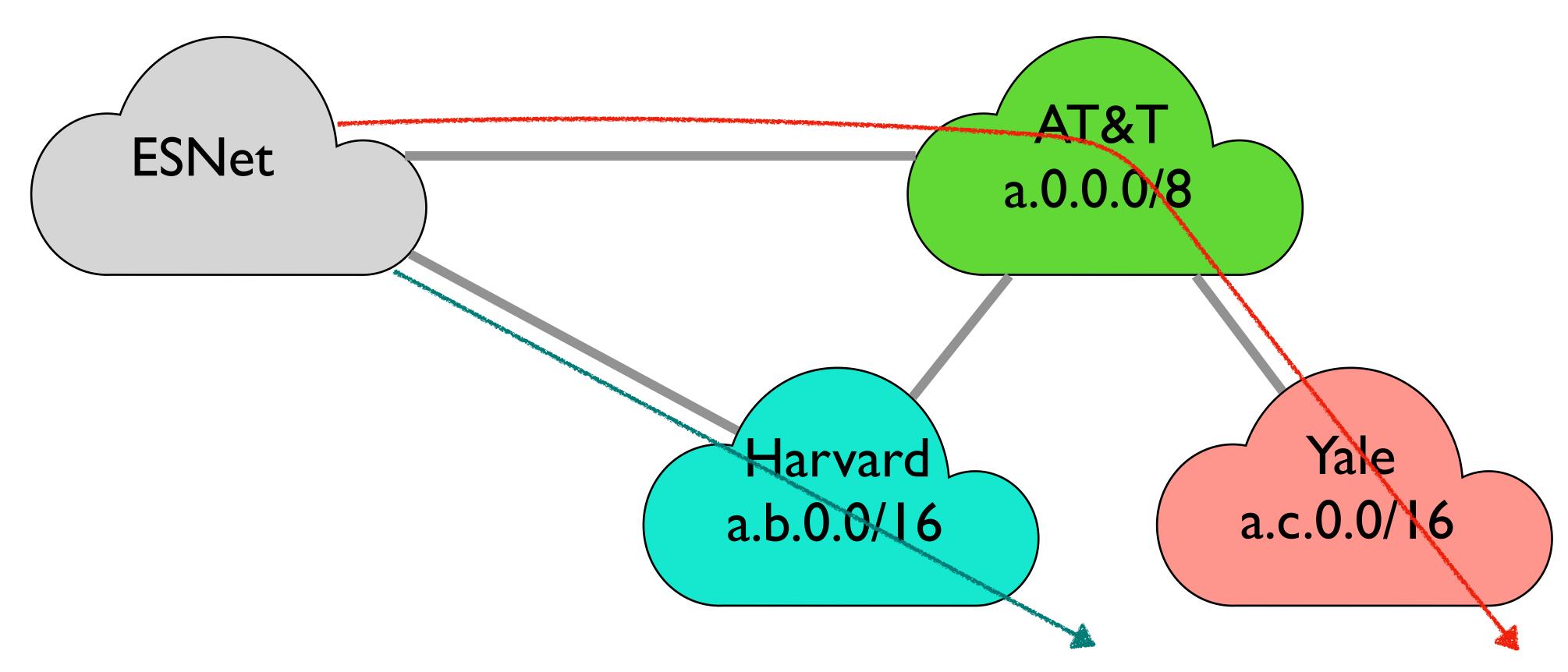
• Problem: May not be able to aggregate addresses for "multi-homed" networks



Harvard is "multi-homed" to AT&T and ESNet (fake example) Multi-homed domain → domain has 2 (or more) providers



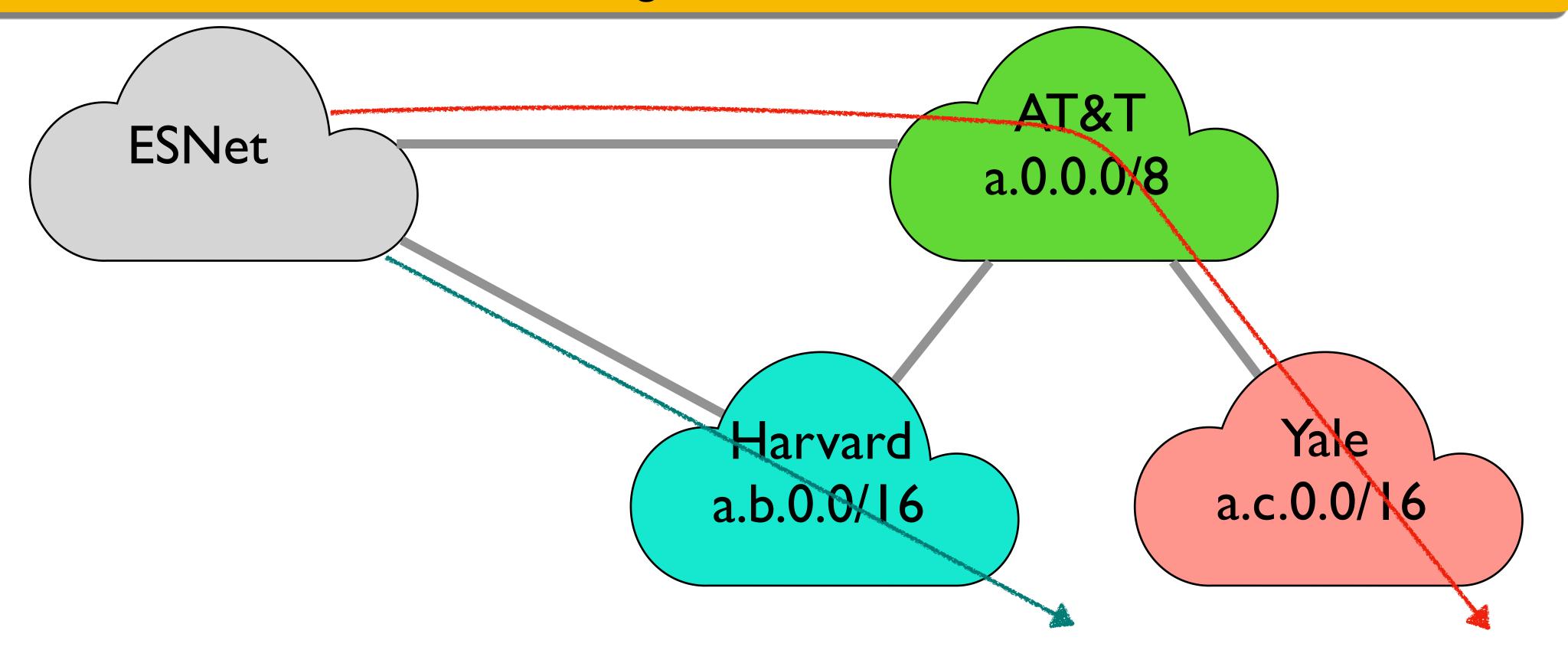
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IP Addressing → Scalable Routing?

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ESNet must maintain routing entries for a.*.*.* and a.b.*.*



IP Addressing → Scalable Routing?

Hierarchical address allocation only helps routing scalability if allocation matches topological hierarchy

- Problem: May not be able to aggregate addresses for "multi-homed" networks
- Two competing forces in scalable routing
 - Aggregation reduces the number of routing entries
 - Multi-homing increases number of entries

BGP & Interdomain Routing

- Destinations are IP prefixes (12.0.0.0/8)
- Nodes are Autonomous Systems (ASes)
- Links represent both physical links and business relationships
 - Customer-provider or peer-to-peer
- Border Gateway Protocol (BGP) is the Interdomain routing protocol
 - Implemented by AS border routers

Topology and Policy is shaped by business relationships between ASes

Three basic kinds of relationships between ASes

- AS A can be AS B's customer
- AS A can be AS B's provider
- AS A can be AS B's peer

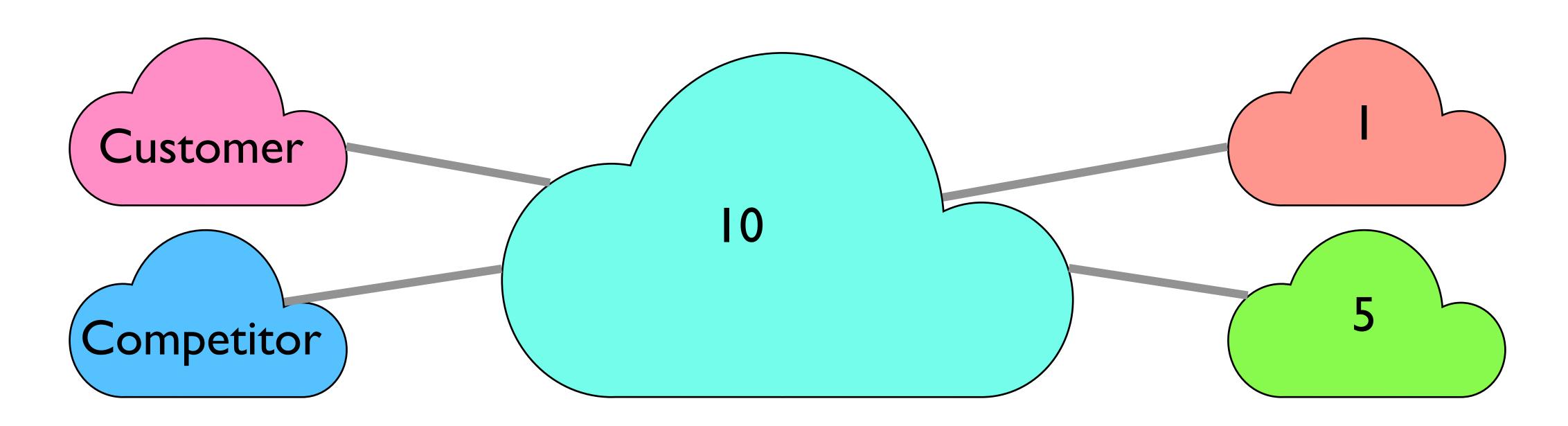
Business implications

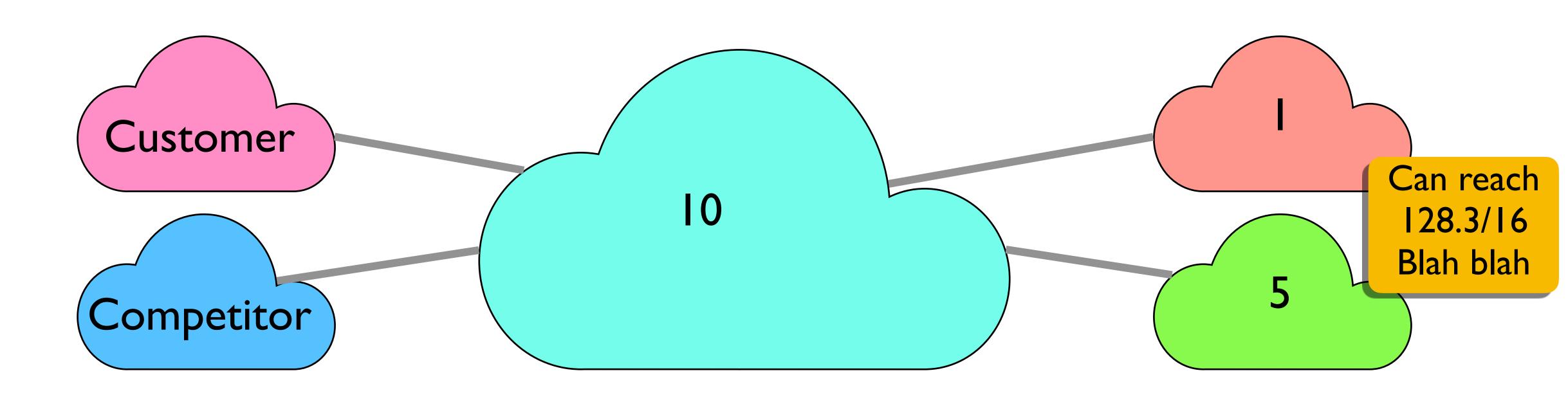
- Customer pays provider
- Peers don't pay each other

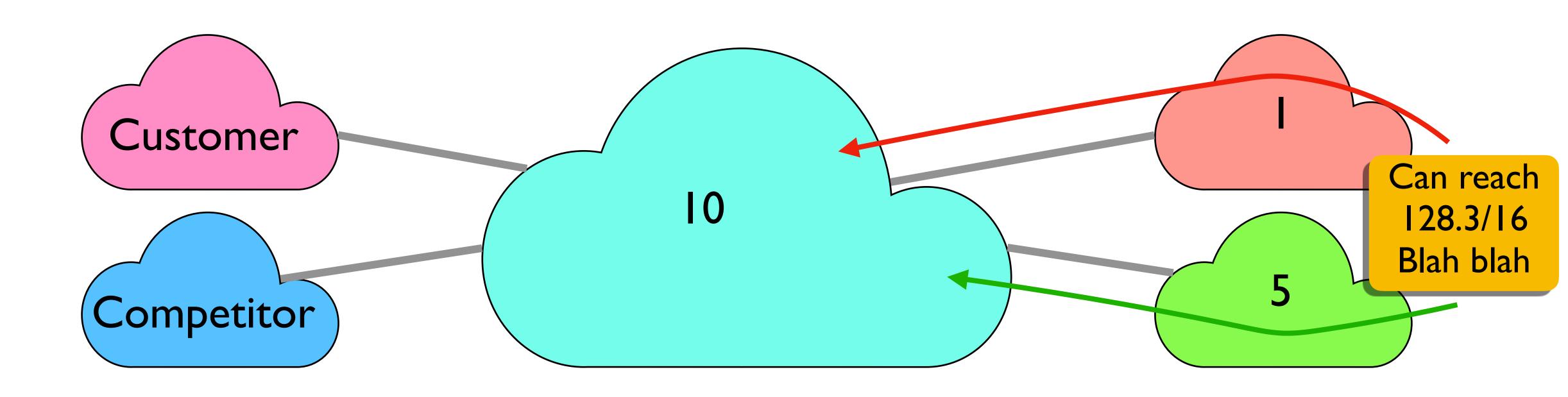
BGP extends DV

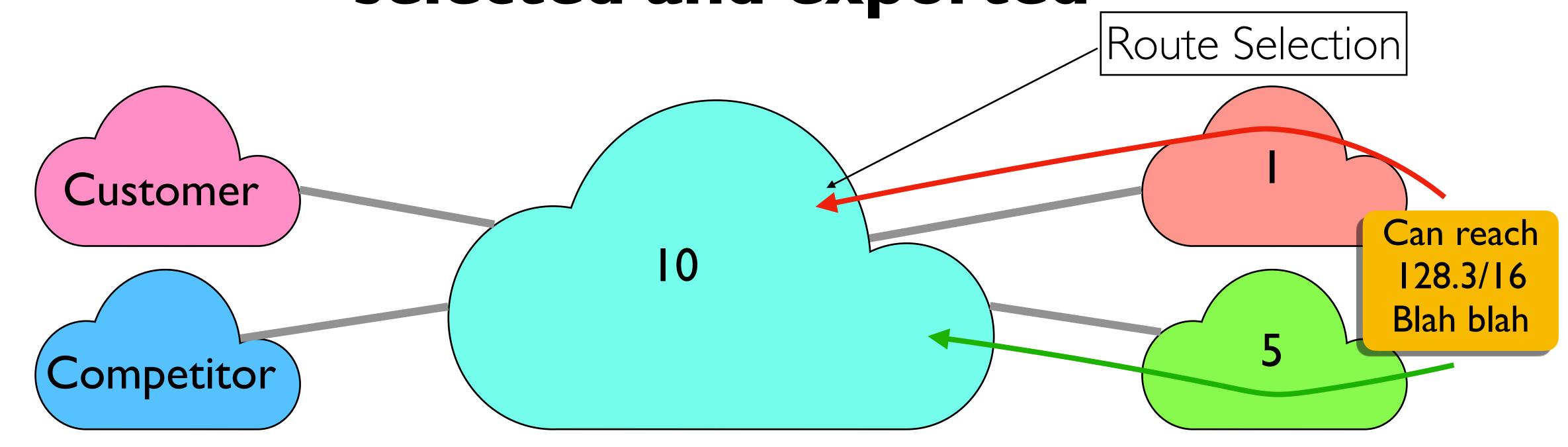
With some important differences

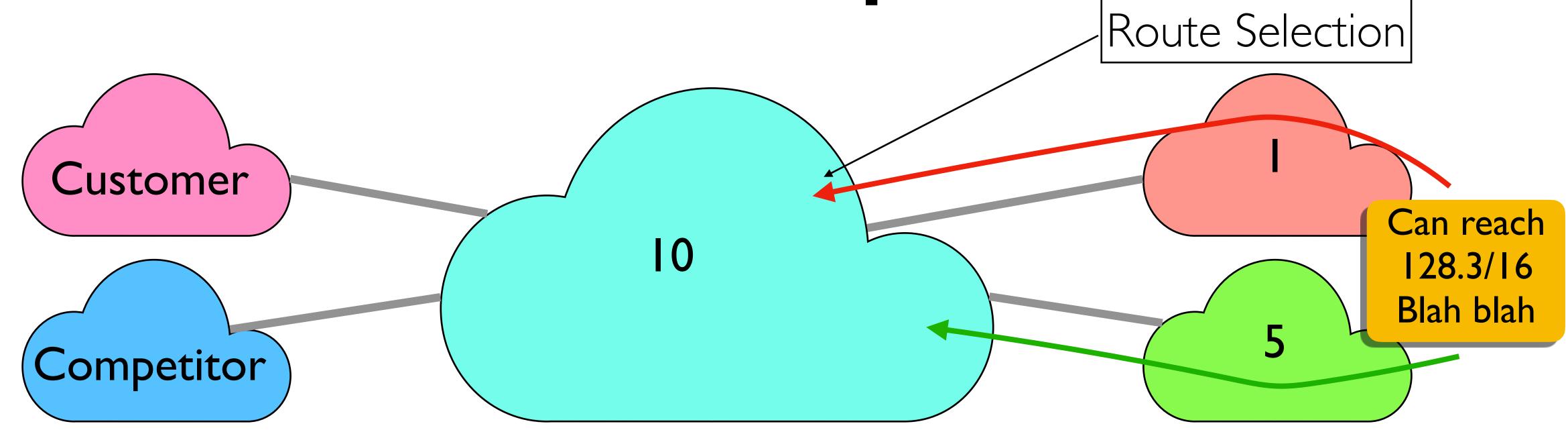
- Routes selected based on policy, not just shorted path
- Path vector (useful to avoid loops)
- Selective route advertisement
- May aggregate routes (aggregating prefixes)



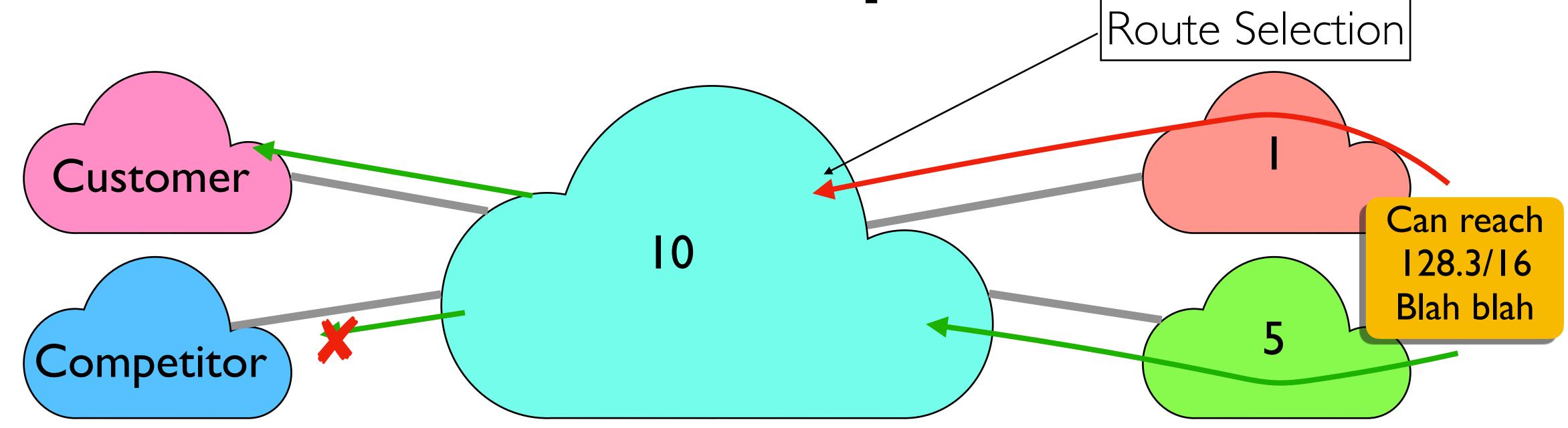




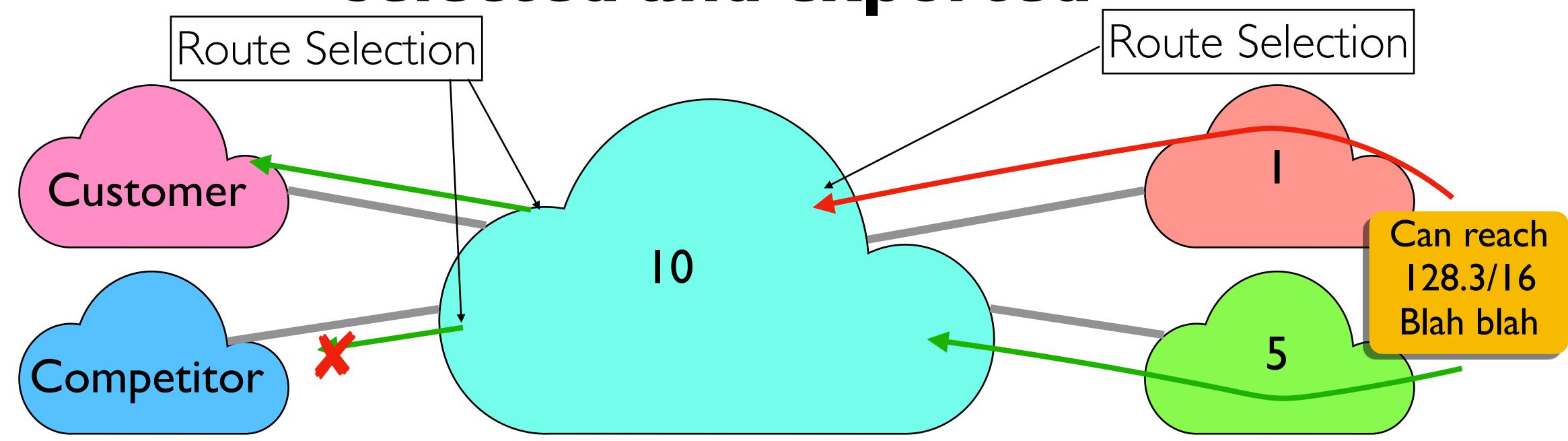




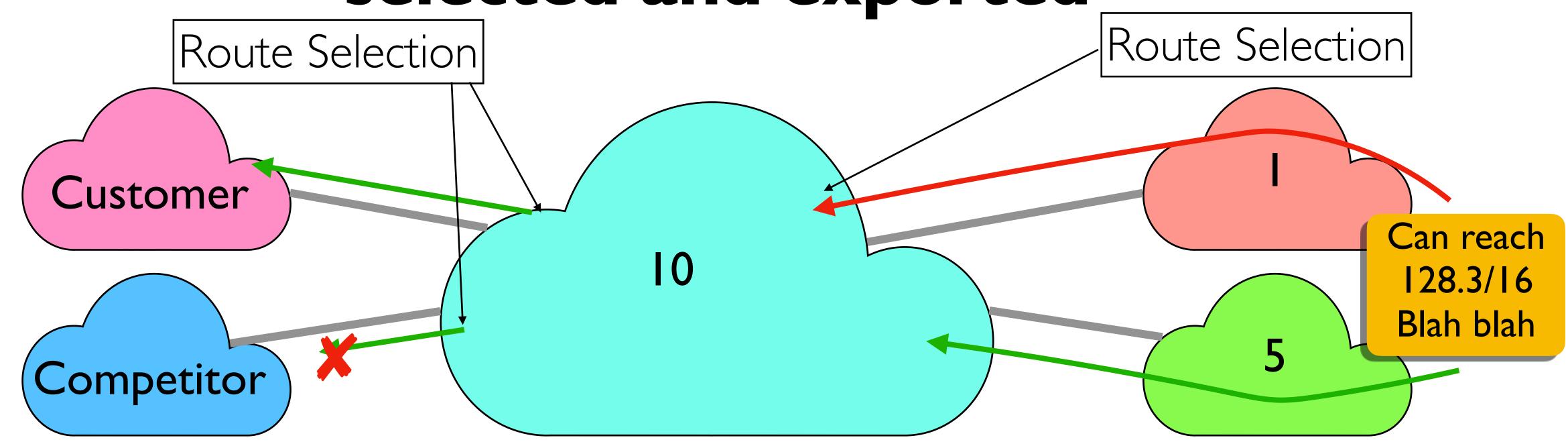
- Selection: Which path to use?
 - Controls whether/how traffic leaves the network



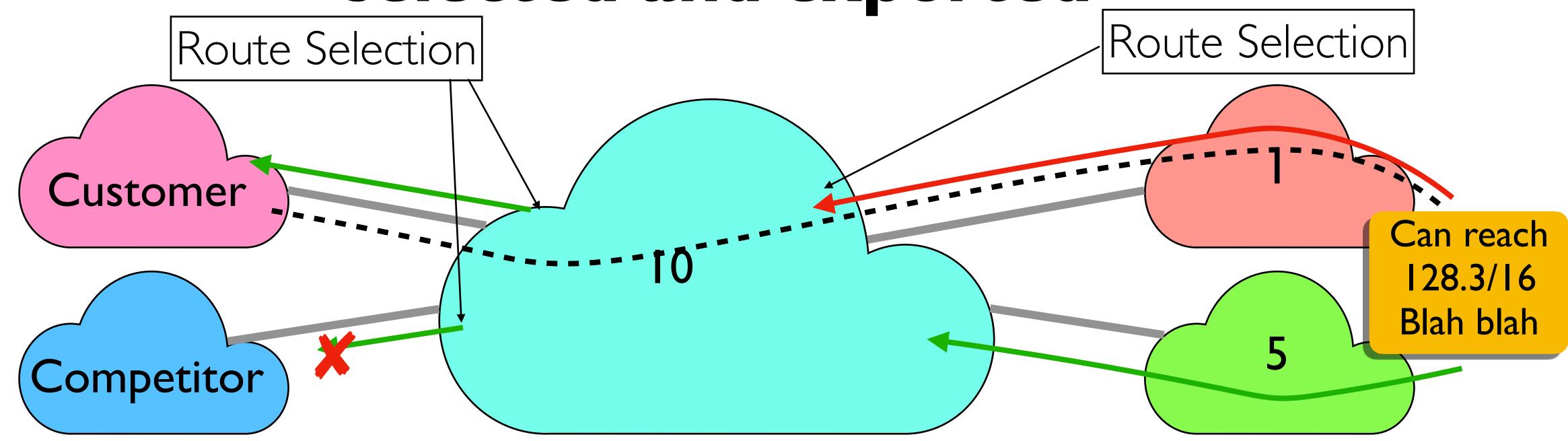
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- Selection: Which path to use?
 - Controls whether/how traffic leaves the network
- Export: Which paths to advertise?
 - Controls whether/how traffic enters the network



- Selection: Which path to use?
 - Controls whether/how traffic leaves the network
- Export: Which paths to advertise?
 - Controls whether/how traffic enters the network

Typical Export Policy

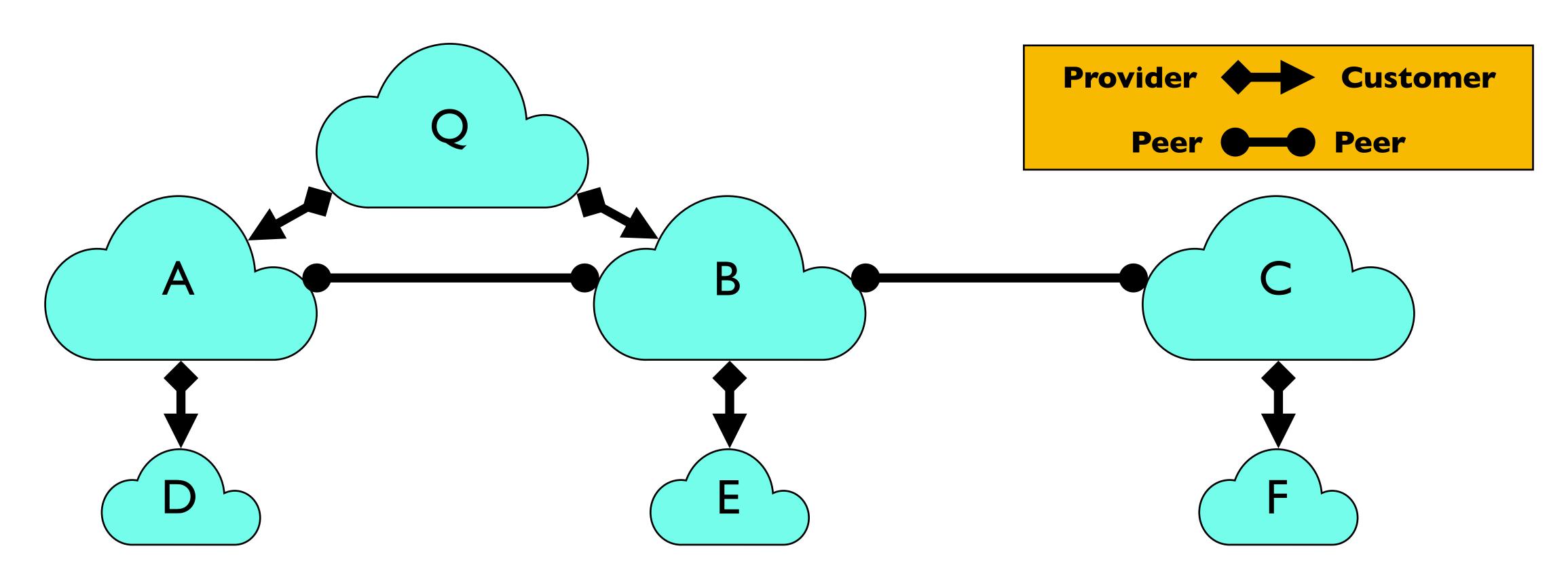
Destination prefix advertised by	Export route to
Customer	Everyone (providers, peers, other customers)
Peer	Customers
Provider	Customers

We'll refer to these as the "Gao-Rexford" rules (Capture common — but not required! — practice)

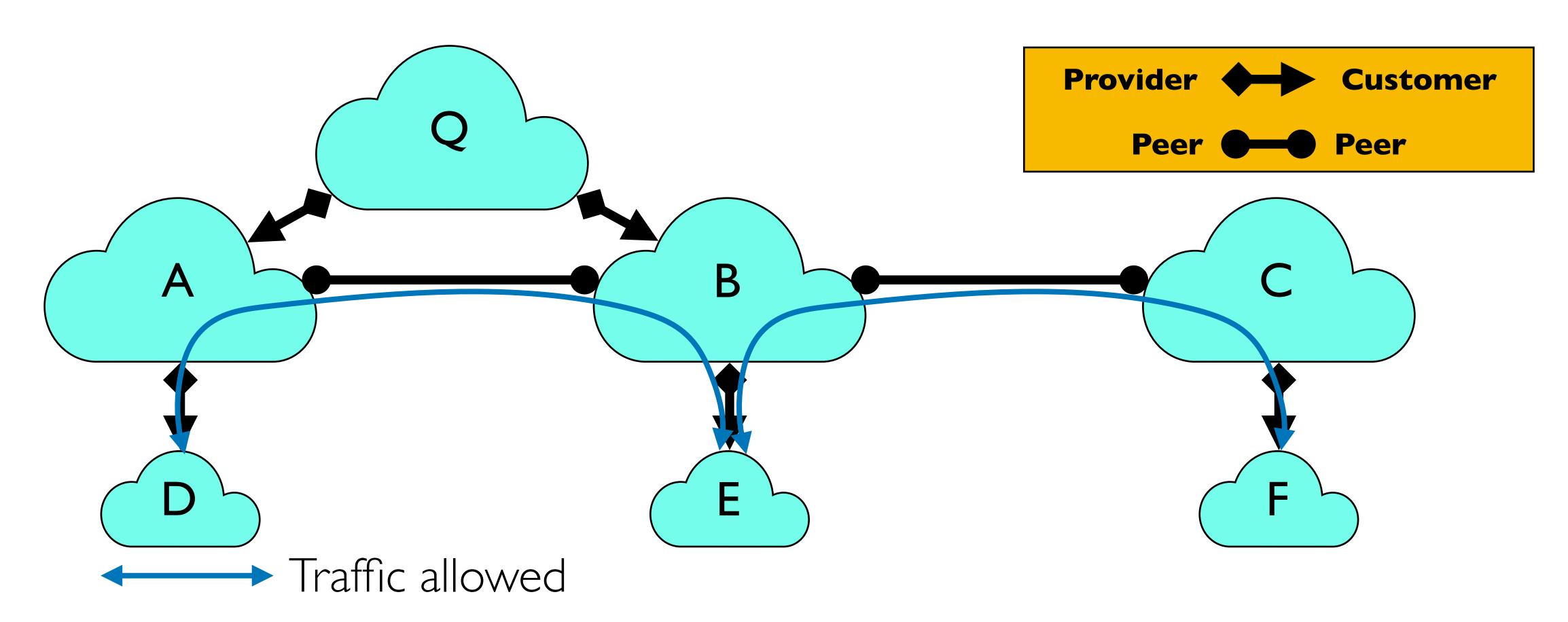
Typical Selection Policy

- In decreasing order of priority
 - Make/save money (send to customer > peer > provider)
 - Maximize performance (small AS path length)
 - Minimize use of my network bandwidth ("hot potato")
 - •
- BGP uses route attributes to implement the above
 - ASPATH, LOCAL_PATH, MED, ...

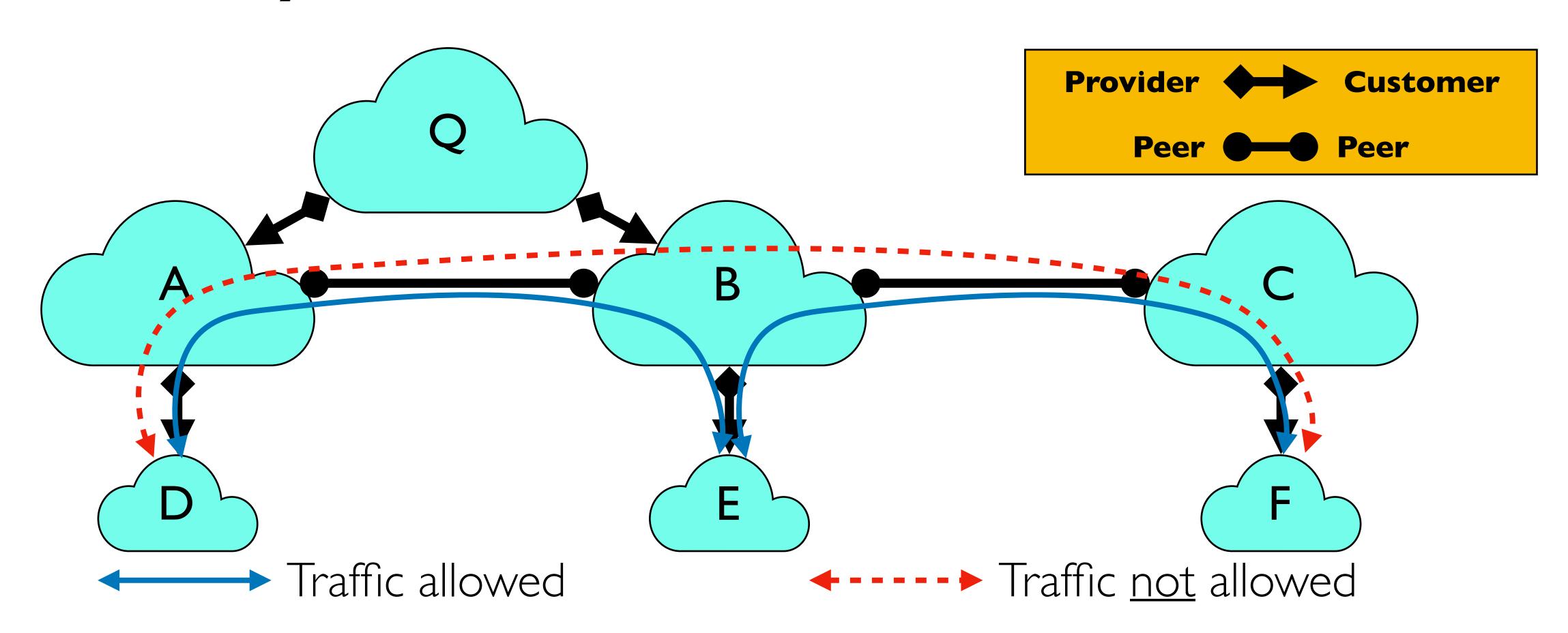
Policy Dictates Route Selection



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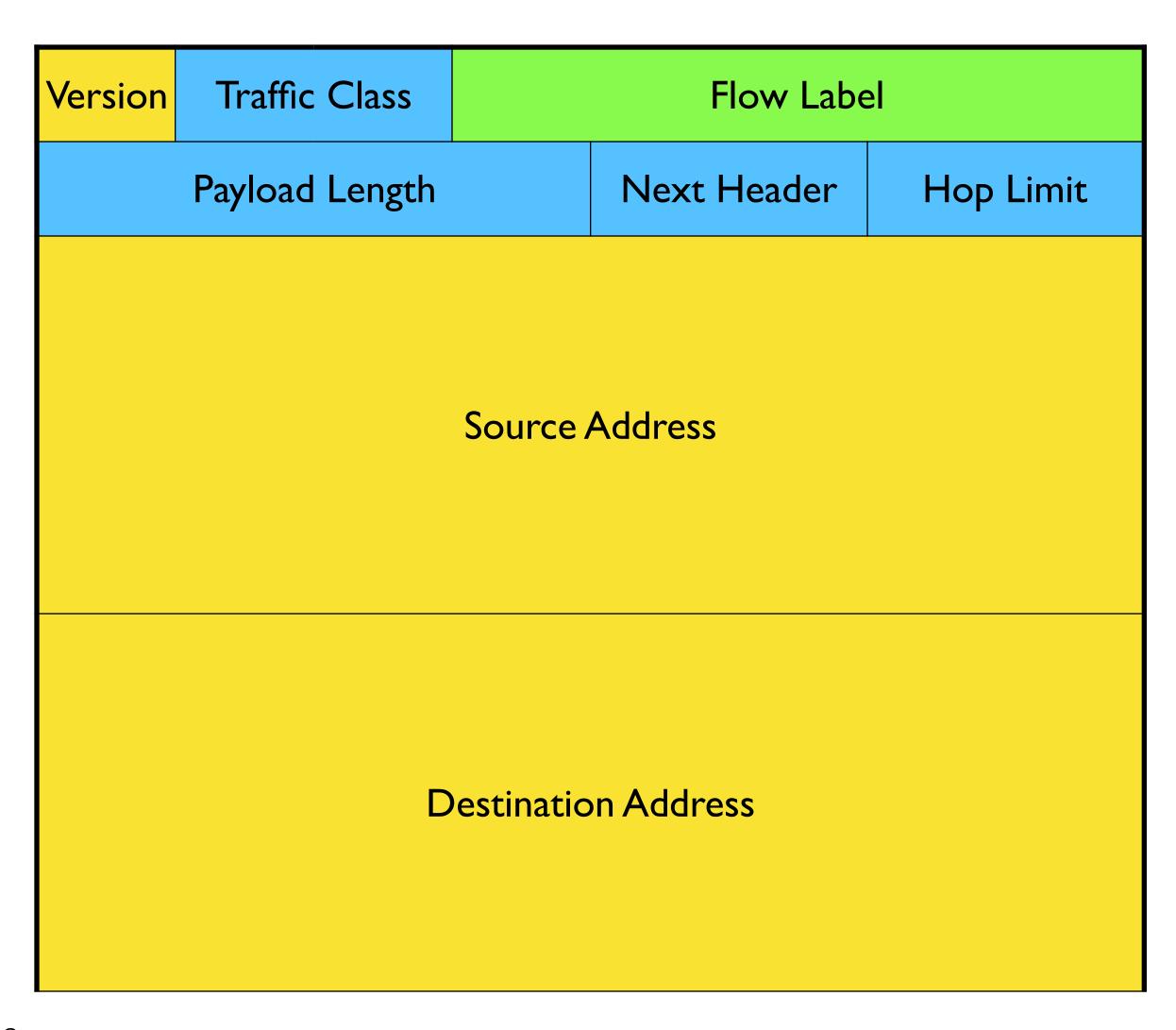
IP Packet Structure

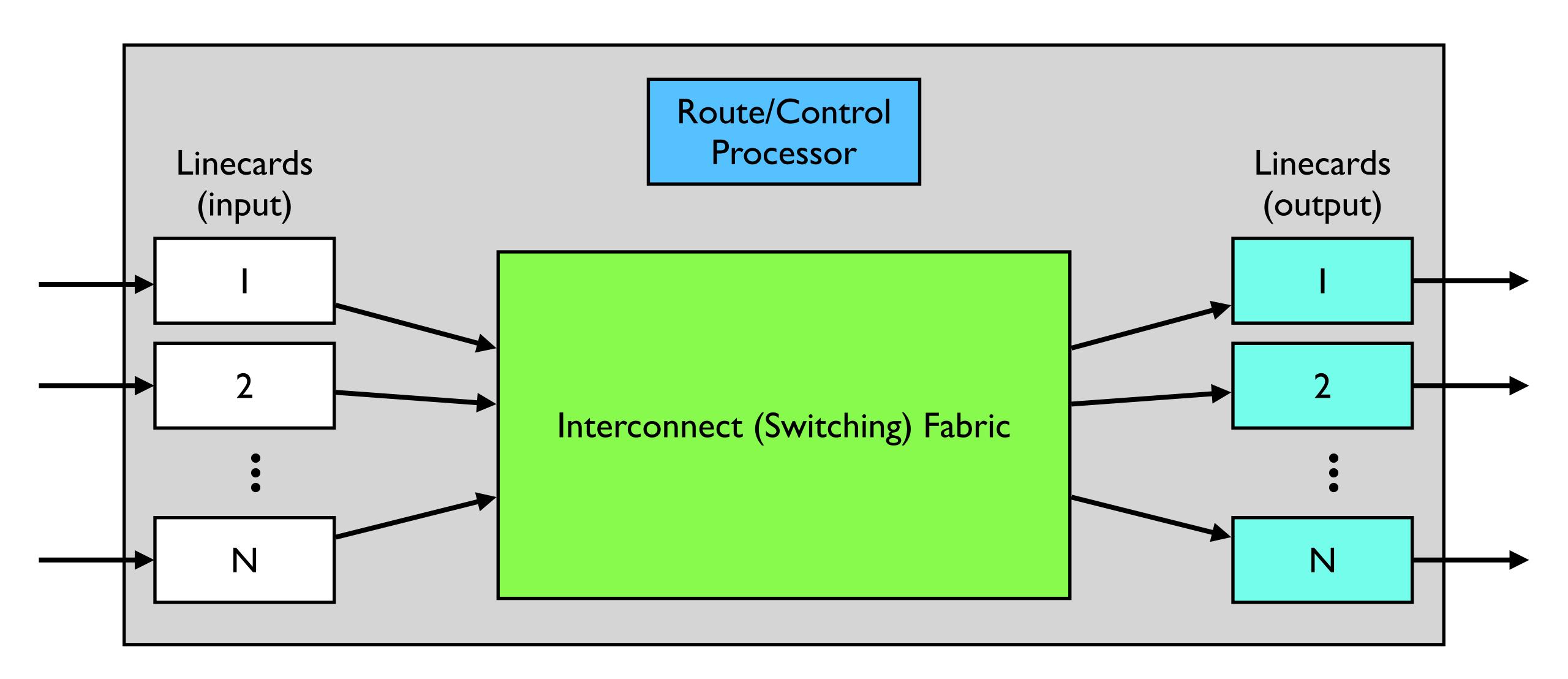
4-bit Version Length	8-bit Type-of-Service	16-bit Total Length (Bytes)			
I6-bit Identification		3-bit Flags	13-bit Fragmentation Offset		
8-bit Time To Live (TTL)	8-bit Protocol	16-bit Header Checksum			
32-bit Source IP Address					
32-bit Destination IP Address					
Options (if any)					
Payload					

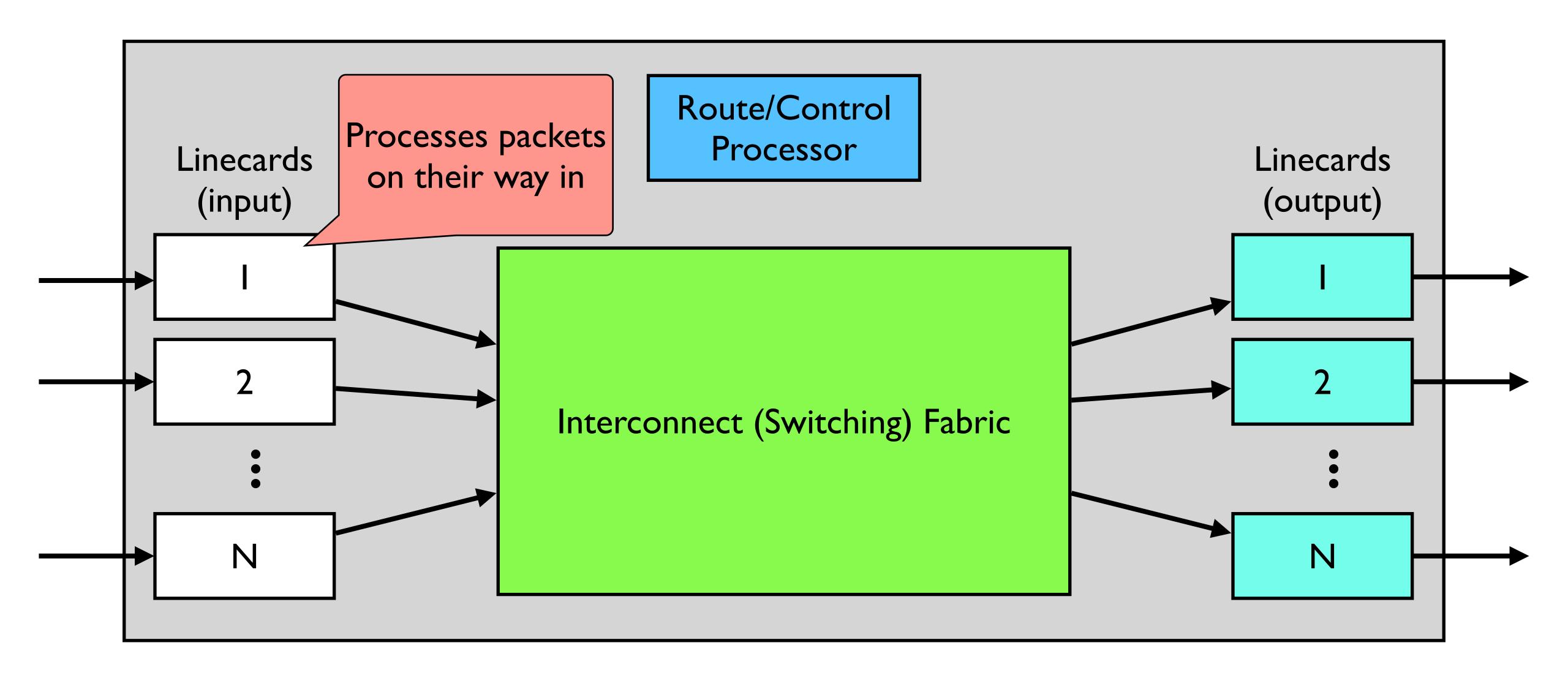
-32 bits —

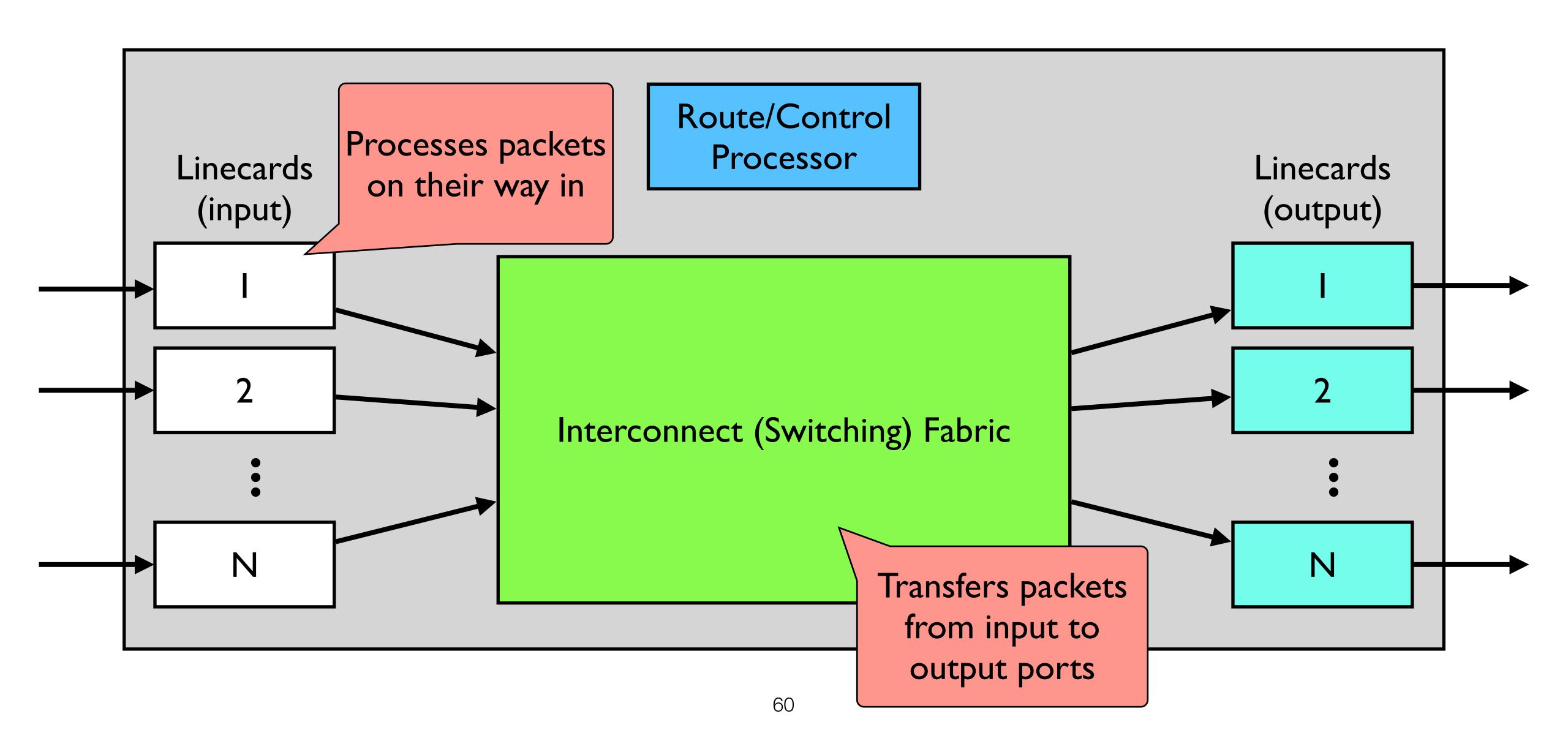
IPv4 and IPv6 Header Comparison

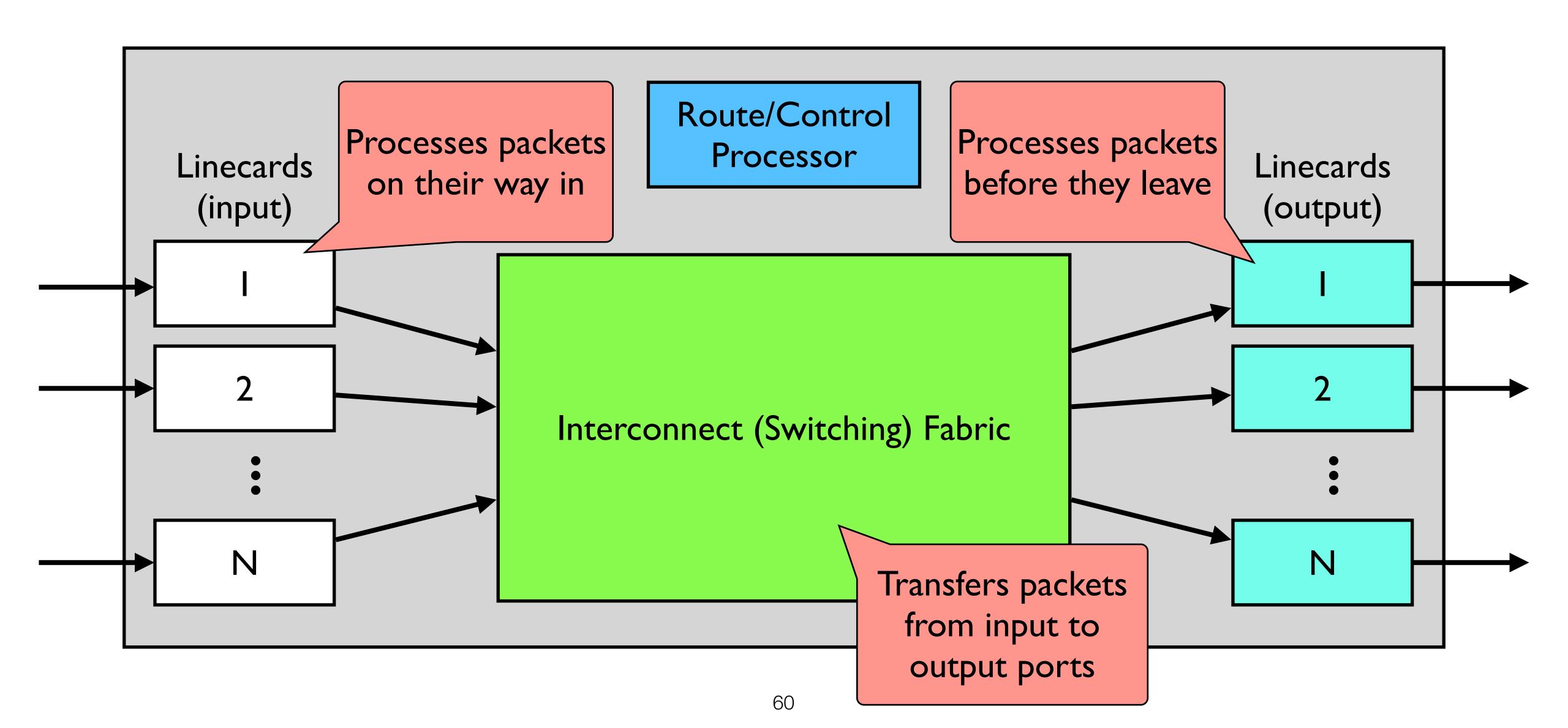
Version	IHL	Type-of-Service	Total Length			
Identification		Flags	Fragmentation Offset			
Time T	o Live	Protocol	Header Checksum			
Source Address						
Destination Address						
Options						
Field name kept from IPv4 to IPv6 Fields not kept in IPv6						
	Field name & position changed in IPv6					
	New field in IPv6					

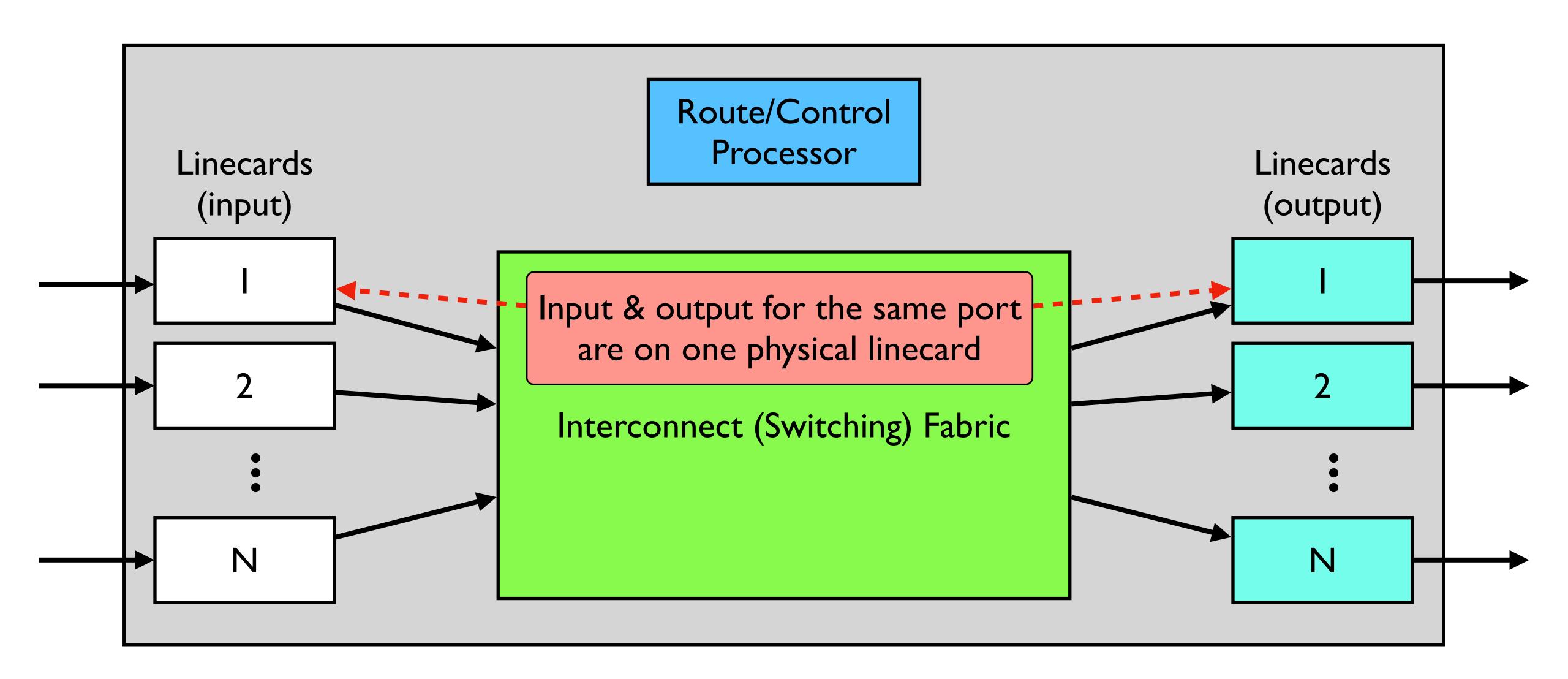


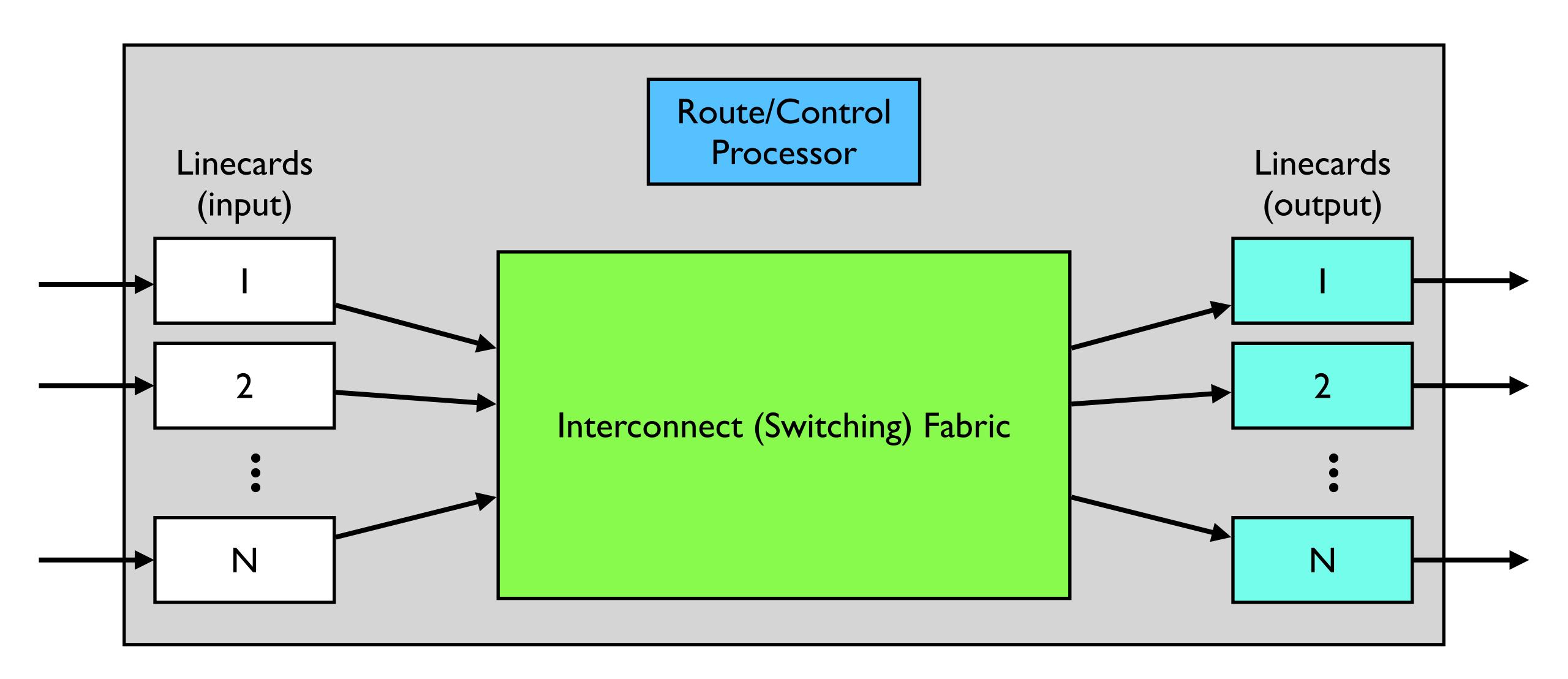


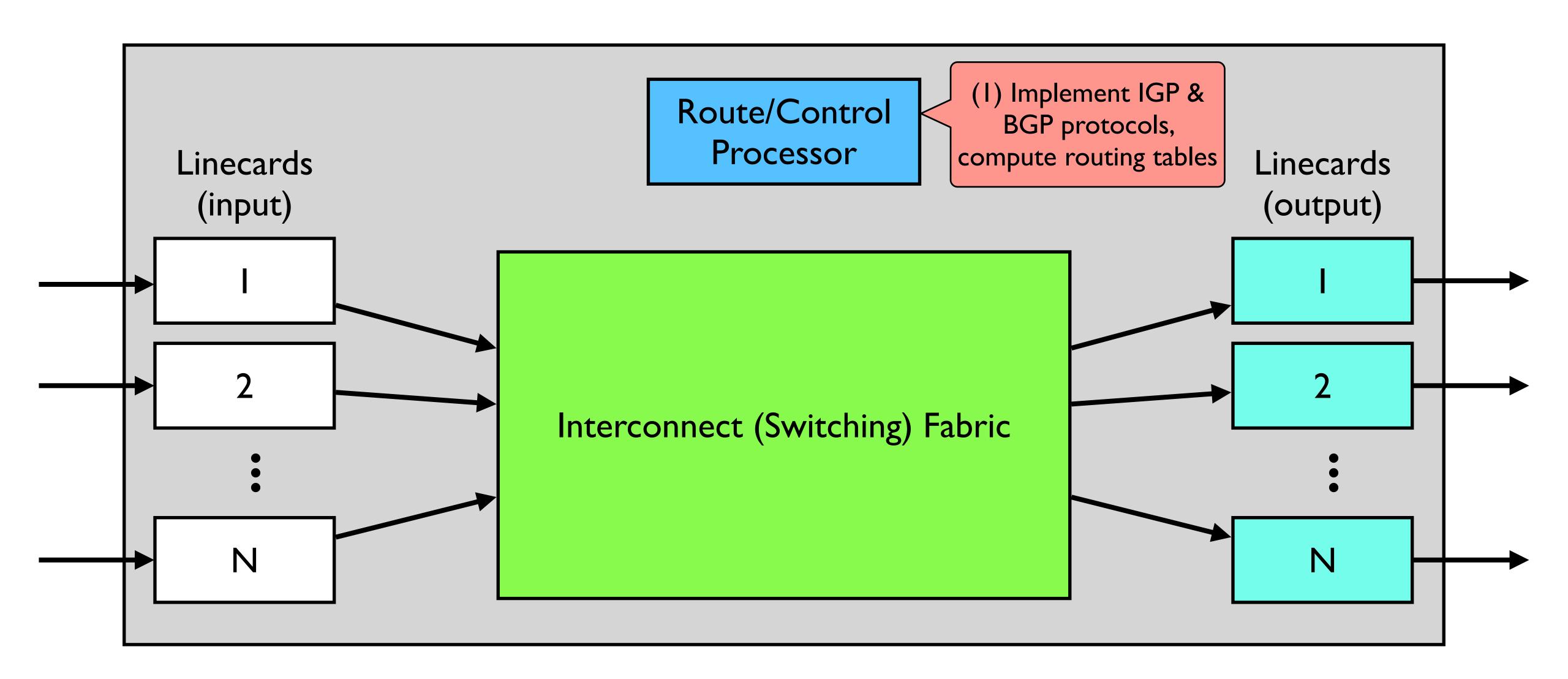


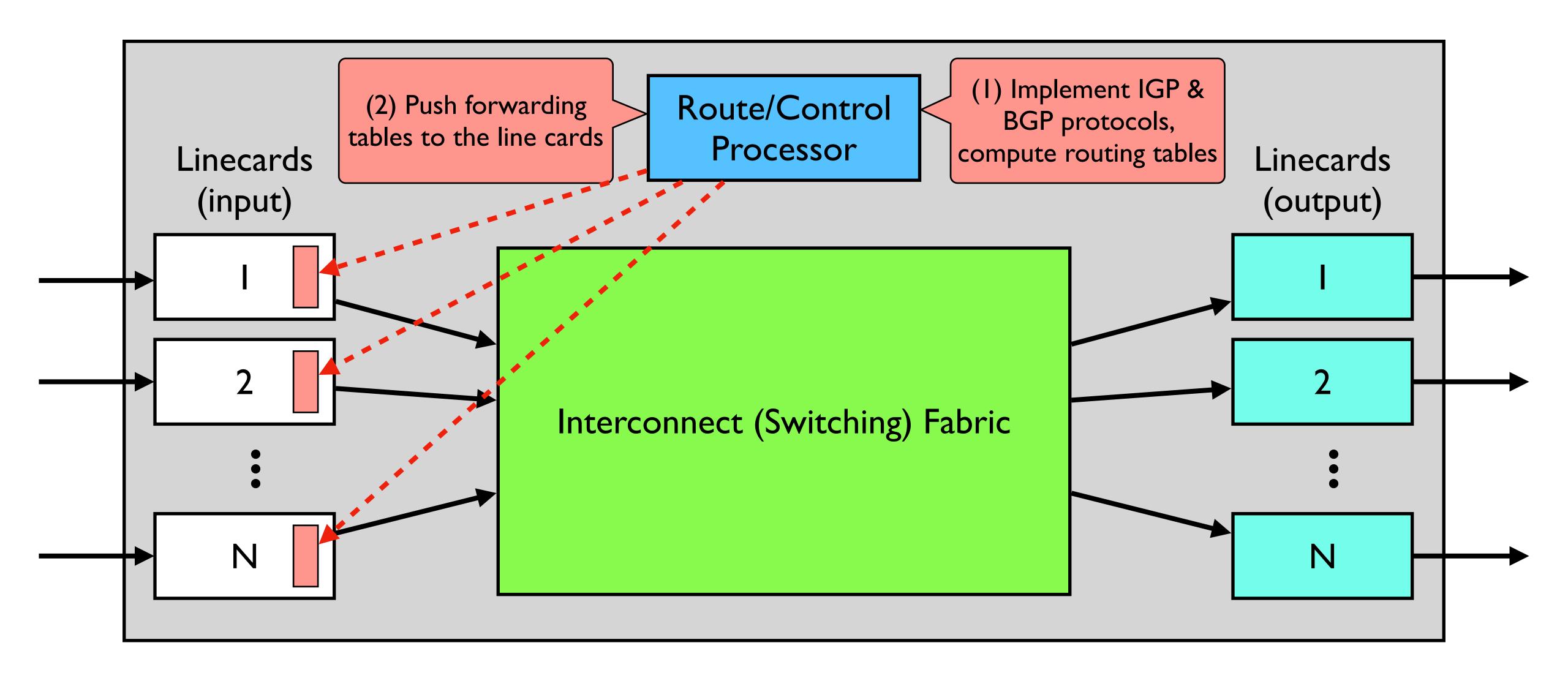


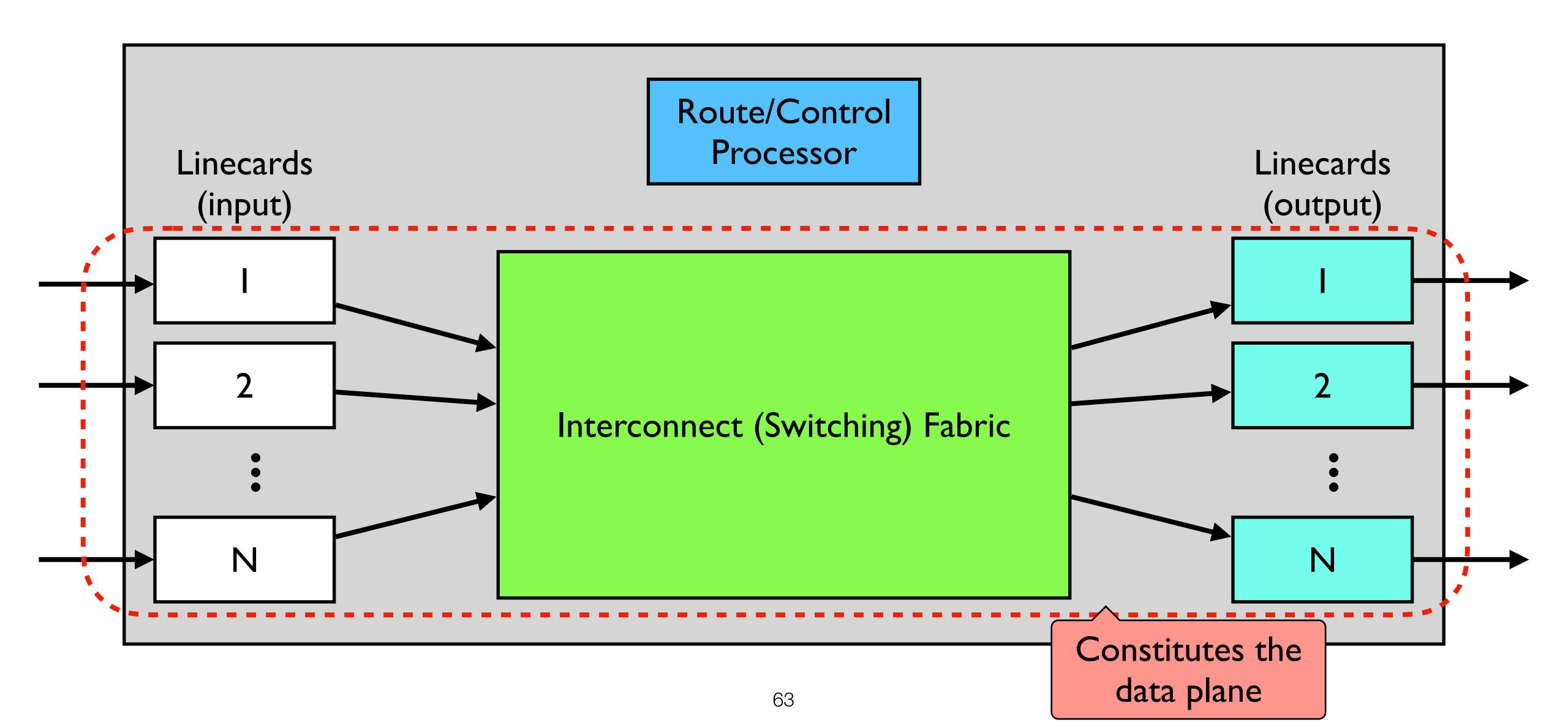


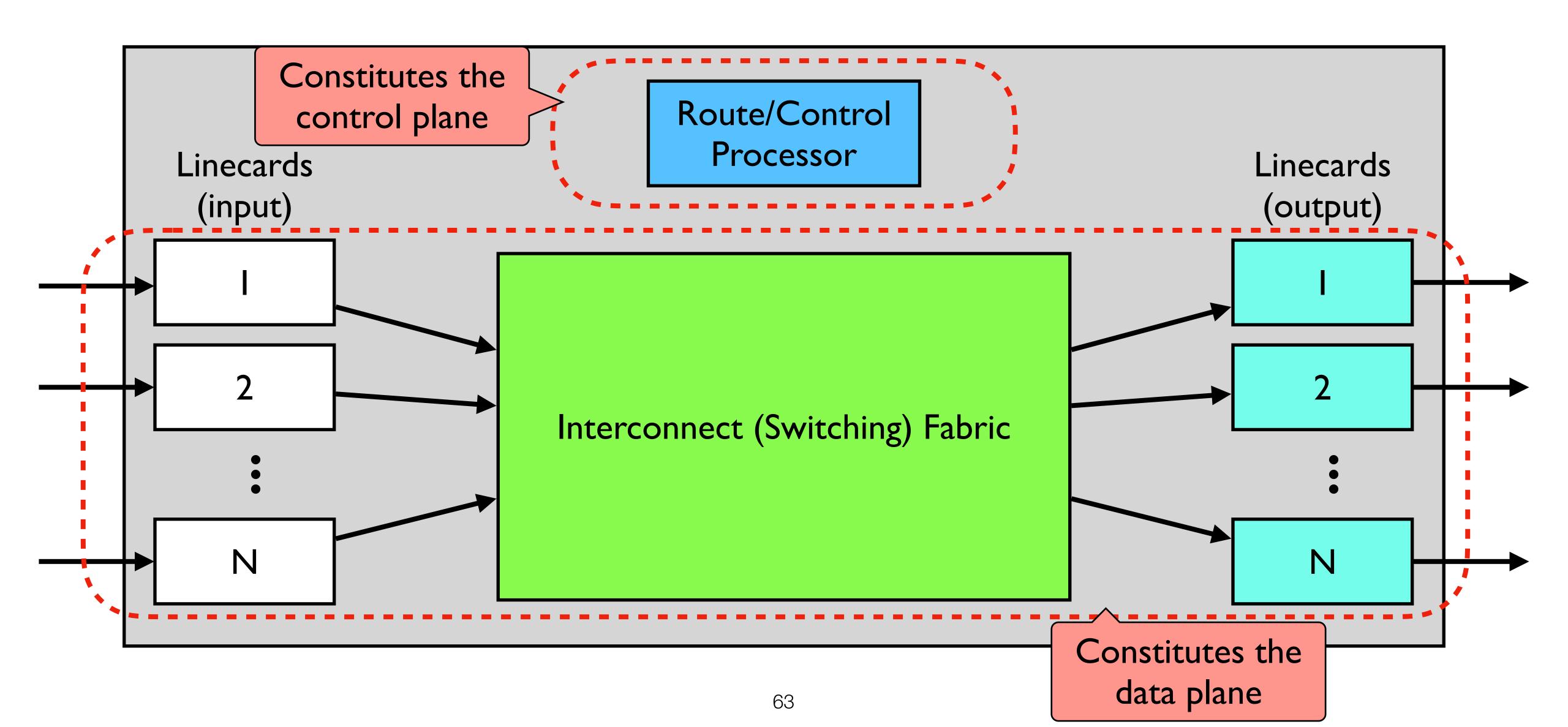












Challenges in Router Design

- @ Line cards: Destination lookups at high speed
 - E.g., find the longest prefix match (LPM) in the table that matches the packet destination address
- @ Switch fabric: head-of-line blocking, scheduling the switch fabric at high speed
- @ Route processor: Complexity/correctness more a problem than performance

Transport Layer

Role of the Transport Layer

Communication between application processes

- Multiplex and demultiplex from/to application processes
- Implemented using **ports** (not the same as router ports!)

Provide common end-to-end services for application layer

- Reliable, in-order data delivery
- Well-placed data delivery

UDP vs. TCP

• Both UDP and TCP provide multiplexing/demultiplexing via ports

	UDP	TCP
Data Abstraction	Packet (datagrams)	Byte-stream of arbitrary length
Service	Best-effort (same as IP)	Reliability, In-order delivery, Flow control, Congestion control
Applications	Video, audio streaming	File transfer, chat

Reliable Transport: General Concepts

- Checksums (for error detection)
- Timers (for loss detection)
- Acknowledgements (feedback from receiver)
 - Cumulative: "Received everything up to X"
 - Selective: "Received X"
- Sequence numbers (detect duplicates, accounting)
- Sliding windows (for efficiency)

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You should know:

- What these concepts are
- Why they exist
- How TCP uses them

Things to know about TCP

- How TCP achieves reliability
- RTT estimation
- Connection establishment/teardown
- Flow Control
- Congestion Control
- Critiques on Congestion Control, Router-based Approaches

• For each, know the how the functionality is implemented, and why it is needed

• Why? TCP uses timeouts to retransmit packets

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- But RTT may vary (significantly!) for different reasons and different timescales
 - Due to temporary congestion
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 - Due to temporary congestion
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- An incorrect RTT estimate might introduce spurious retransmissions or overly long delays
- RTT estimators should react to change, but not too quickly
 - Proposed solutions used EWMA, incorporate deviations

E.g., Reliability

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- Why? IP is best-effort but many applications need reliable delivery
 - Having TCP take care of it simplifies application development!

E.g., Reliability

- Why? IP is best-effort but many applications need reliable delivery
 - Having TCP take care of it simplifies application development!

• How?

- Checksums & timers (for error and loss detection)
- Fast retransmit (for faster-than-timeout loss detection)
- Cumulative ACKs (feedback from receiver what's lost/what's not)
- Sliding windows (for efficiency)
- Buffers at sender (to hold packets while waiting for ACKs)
- Buffers at receiver (to reorder packets before delivery to application)

E.g., Connection Establishment

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• Why?

- TCP is a stateful protocol (CWND, buffer-space, ISN, etc.)
- Need to initialize connection state at both ends
- Exchange initial sequence numbers

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- Need to initialize connection state at both ends
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How? Three-way handshake

- Host A sends a SYN to host B
- Host B returns a SYN acknowledgement (SYN ACK)
- Host A sends an ACK (+data) to acknowledge the SYN ACK
- Hosts exchange proposed Initial Sequence Numbers at each step

E.g., Flow Control

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• Why?

- TCP offers a reliable in-order byte-stream abstraction
- Hence, TCP at the receiver must buffer a packet until all packets before it (in byte-order) have arrived and the receiving application has consumed available bytes
- Hence, receiver advances its window when the receiving application consumes data
- But, sender advances its window when new data is ACKed
- Hence, risk the sender might overrun the receiver's buffers

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• How? "Advertised Window" field in TCP header

- Receiver advertises the "right hand edge" of its window to sender
- Sender agrees not to exceed the amount

E.g., Congestion Control

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• Why?

- Because a sender shouldn't overload the network itself
- But, should make efficient use of available network capacity
- While sharing the available capacity fairly with other flows
- And adapting to changes in available capacity

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• Why?

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• How?

- Quickly find current available capacity (slow start)
- Then adapt to changes (congestion avoidance, AIMD)
- With optimizations (fast recovery)
- Study the TCP state machine diagram from the text!

- TCP Throughput Equation
 - Know the equation
 - Understand its implications

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$$= \sqrt{\frac{3}{2}} \cdot \frac{1}{RTT \cdot \sqrt{p}}$$

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TCP Throughput Equation

- Know the equation
- Understand its implications
- Max-min fairness / fair queuing
 - Definition, know how to calculate fair-rate
 - Know pros and cons of FQ (isolation, router complexity)
- Router assisted congestion control
 - Pros (better info.) and cons (complexity in routers)
 - E.g. explicit rate allocation (RCP) and explicit congestion notification (ECN)

Final Questions?

Good luck!