

# On the Characteristics and Origins of Internet Flow Rates

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# Motivation

- Limited knowledge about flow rates
  - Flow rates are impacted by many factors
    - Congestion, bandwidth, applications, host limits, ...
  - Little is known about the resulting rates or their causes
- Why is it important to understand flow rates?
  - Understanding the network
    - User experience
  - Improving the network
    - Identify and eliminate bottlenecks
  - Designing scalable network control algorithms
    - Scalability depends on the distribution of flow rates
  - Deriving better models of Internet traffic
    - Useful for workload generation and various network problems

# Two Questions

- What are the characteristics of flow rates?
  - Rate distribution
  - Correlations
- What are the causes of flow rates?
  - T-RAT: TCP Rate Analysis Tool
    - Design
    - Validation
    - Results

# **Characteristics of Internet Flow Rates**

# Datasets and Methodology

- Datasets
  - Packet traces at ISP backbones and campus access links
    - 8 datasets; each lasts 0.5 – 24 hours; over 110 million packets
  - Summary flow statistics collected at 19 backbone routers
    - 76 datasets; each lasts 24 hours; over 20 billion packets
- Flow definition
  - Flow ID: <SrcIP, DstIP, SrcPort, DstPort, Protocol>
  - Timeout: 60 seconds
- Rate = Size / Duration
  - Exclude flows with duration < 100 msec
- Look at:
  - Rate distribution
  - Correlations among rate, size, and duration

# Flow Rate Characteristics

- Rate distribution
  - Most flows are *slow*, but most bytes are in *fast* flows
  - Distribution is skewed
    - Not as skewed as size distribution
    - Consistent with log-normal distribution [BSSK97]
- Correlations
  - Rate and size are strongly correlated
  - Not due to TCP slow-start
    - Removed initial 1 second of each connection; correlations increase
  - What users download is a function of their bandwidth

# **Causes of Internet Flow Rates**

# T-RAT: TCP Rate Analysis Tool

- Goal
  - Analyze TCP packet traces and determine rate-limiting factors for different connections
- Requirements
  - Work for traces recorded anywhere along a network path
    - Traces don't have to be recorded near an endpoint
  - Work just seeing one direction of a connection
    - Data only or ACK only → there is no easy cause & effect
  - Work with partial connections
    - Prevent bias against long-lived flows
  - Work in a streaming fashion
    - Avoid having to read the entire trace into memory



# TCP Rate Limiting Factors

<b><i>Factor</i></b>	<b><i>Description</i></b>
<i>Application</i>	Application doesn't generate data fast enough
<i>Opportunity</i>	Flow never leaves slow-start
<i>Receiver</i>	Flow is limited by receiver advertised window
<i>Sender</i>	Flow is limited by sending buffer
<i>Bandwidth</i>	Flow fully utilizes and is limited by bottleneck link bandwidth
<i>Congestion</i>	Flow responds to packet loss
<i>Transport</i>	Flow has left slow-start & no packet loss & not limited by receiver/sender/bandwidth

# T-RAT Components

- MSS Estimator
  - Identify *Maximum Segment Size (MSS)*
- RTT Estimator
  - Estimate RTT
  - Group packets into flights
    - Flight: packets sent during the same RTT
- Rate Limit Analyzer
  - Determine rate-limiting factors based on MSS, RTT, and the evolution of flight size

# What Makes It Difficult?

- The network may introduce a lot of noise
  - E.g. significant delay variation, ACK compression, ...
- Time-varying RTT is difficult to track
  - E.g., handshake delay and median RTT may differ substantially
- Delayed ACK significantly complicates TCP dynamics
  - E.g. congestion avoidance: 12, 12, 13, 12, 12, 14, 14, 15, ...
- There are a large number of TCP flavors & implementations
  - Different loss recovery algorithms, initial cwnd, bugs, weirdness
- Timers may introduce behavior difficult to analyze
  - E.g. delack timer may expire in the middle of an RTT
- Packets missing due to packet filter drop, route change
  - They are not lost!
- There may be multiple causes for a connection
- Some behaviors are intrinsically ambiguous
- *And a lot more ...*

# MSS Estimator

- Data stream
  - MSS ← largest data packet payload
- ACK stream
  - MSS ← “*most frequent common divisor*”
    - Like GCD, apply heuristics to
      - avoid looking for divisors of numbers that are not multiples of MSS
      - favor popular MSS (e.g. 536, 1460, 512)

# RTT Estimator

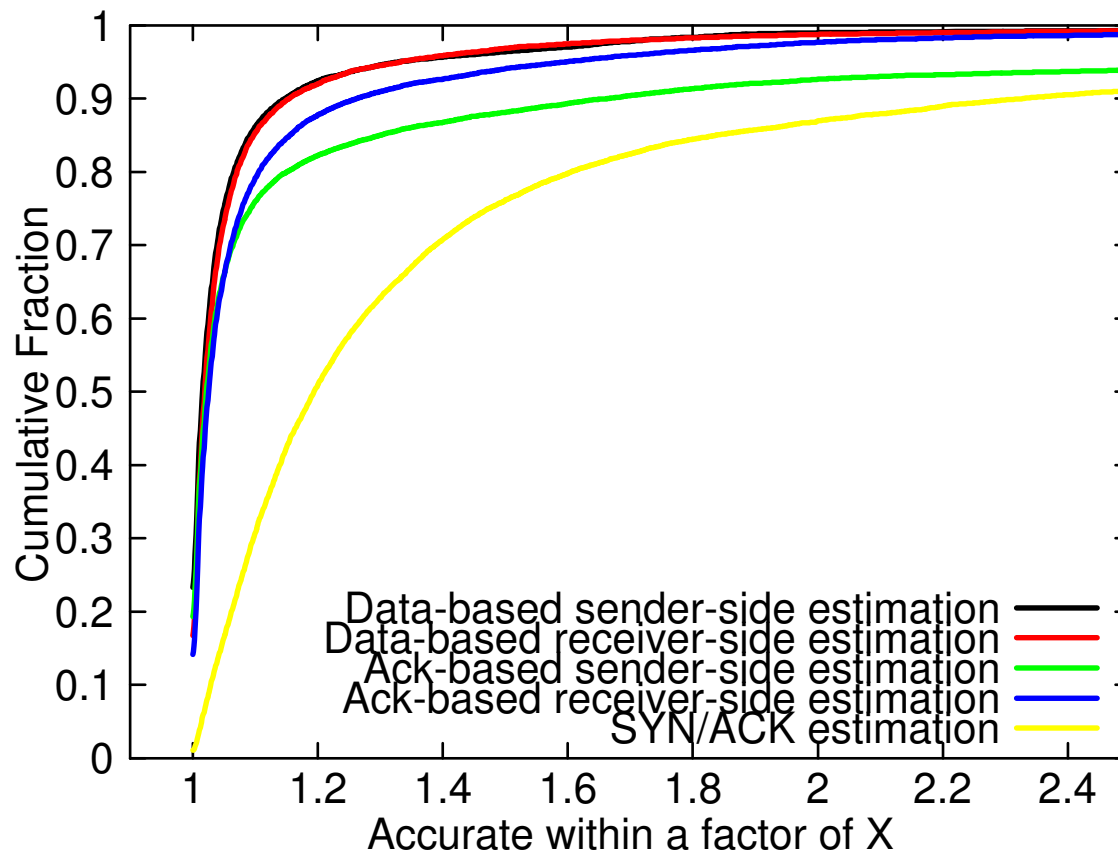
- Generate a set of candidate RTTs
  - Between 3 msec and 3 sec:  $0.003 \times 1.3^K$  sec
- Assign a score to each candidate RTT
  - Group packets into flights
    - Flight boundary: packet with large inter-arrival time
  - Track evolution of flight size over time and match it to identifiable TCP behavior
    - Slow start
    - Congestion avoidance
    - Loss recovery
  - Score  $\leftarrow$  # packets in flights consistent with identifiable TCP behavior
- Pick the top scoring candidate RTT

# Rate Limit Analyzer

<b>Cause</b>	<b>Test</b>
<i>Application</i>	A sub-MSS packet followed by a lull $>$ RTT, followed by new data
<i>Opportunity</i>	Total bytes $<$ $13 * \text{MSS}$ , or it never leaves slow-start
<i>Receiver</i>	3 consecutive flights with size $>$ $\text{awnd}_{\text{max}} - 3 * \text{MSS}$
<i>Sender</i>	Flight size stays constant, and not receiver limited
<i>Bandwidth</i>	A flow keeps achieving same amount of data in flights prior to loss; or it has nearly equally spaced packets
<i>Congestion</i>	Packet loss and not bandwidth-limited
<i>Transport</i>	Sender has left slow-start and there is no packet loss
<i>Host</i>	Flight size stays constant, but we only see data packets and can't tell between sender and receiver
<i>Unknown</i>	None of the above

# RTT Validation

Validation against *tcpanaly* [Pax97] over NPD  $N_2$  (17,248 conn)



**RTT estimator works reasonably well in most cases**

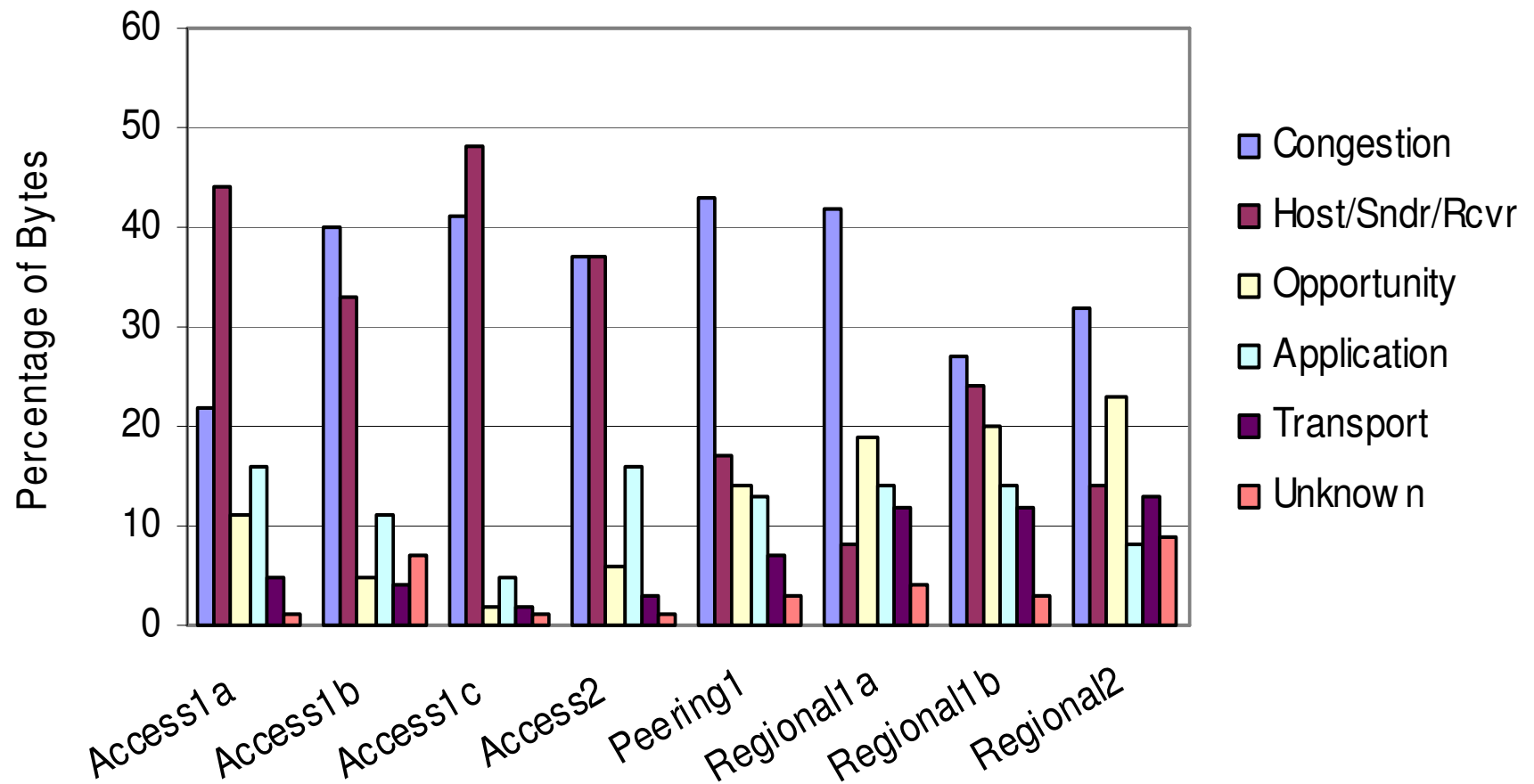
# Rate Limit Validation

- Methodology
  - ns2 simulations + dummynet experiments
- T-RAT correctly identifies the cause in vast majority of cases
- Failure scenarios

<i>Rcvr/Sndr</i>	Window = 2 packets; or link utilization > 90%
<i>Transport</i>	High background load (esp. on ACK stream)
<i>Congestion</i>	“transport limited” w/o loss → NOT failure!
<i>Bandwidth</i>	RTT wrong, but rate limit correct
<i>Opportunity</i>	Connection size is very large
<i>Application</i>	Less accurate with Nagle’s algorithm

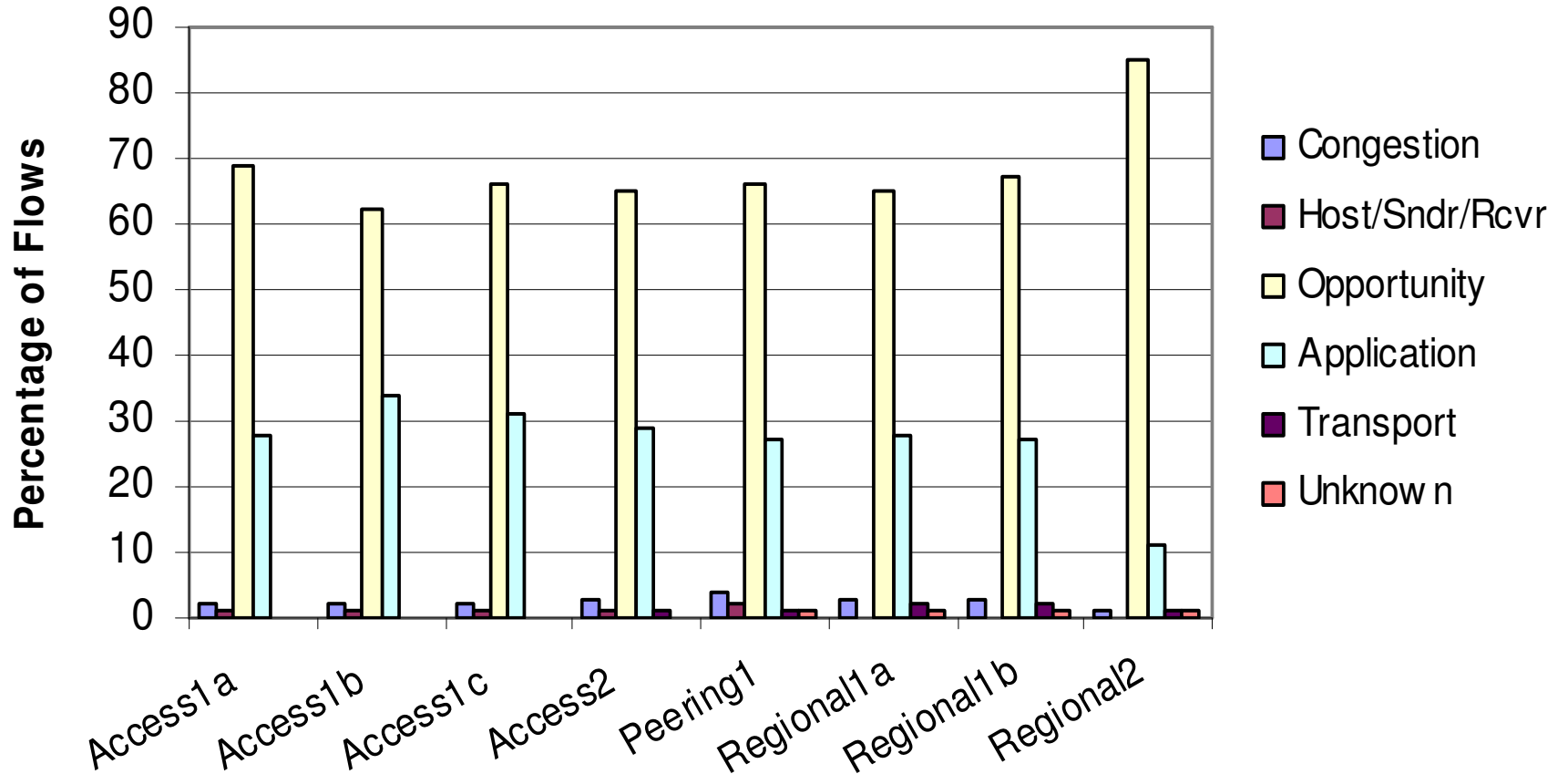


# Rate Limiting Factors (Bytes)



**Dominant causes by bytes: Congestion, Receiver**

# Rate Limiting Factors (Flows)



**Dominant causes by flows: Opportunity, Application**

# Flow Characteristics by Cause

- Different causes are associated with different performance for users
  - Rate distribution
    - Highest rates: Receiver, Transport
  - Size distribution
    - Largest sizes: Receiver
  - Duration distribution
    - Longest duration: Congestion

# Conclusion

- Characteristics of Internet flow rates
  - Fast flows carry most of the bytes
    - It is important to understand their behavior.
  - Strong correlation between flow rate and size
    - What users download is a function of their bandwidth.
- Causes of Internet flow rates
  - Dominant causes:
    - In terms of bytes: congestion, receiver
    - In terms of flows: opportunity, application
  - Different causes are associated with different performance
- T-RAT has applicability beyond the results we have so far
  - E.g. correlating rate limiting factors with other user characteristics like application type, access method, etc.

# Thank you!

<http://www.research.att.com/projects/T-RAT/>