

# Understanding the **Processing Delay** of Peer-to-Peer Relay Nodes

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# Motivation

Peer-to-peer relaying is commonly used

- Purpose

- overlay routing (for better quality)
- connection establishment (for peers behind NAT/firewall)
- data aggregation (lower bandwidth usage)

- Application

- Conferencing:
- Video Streaming:

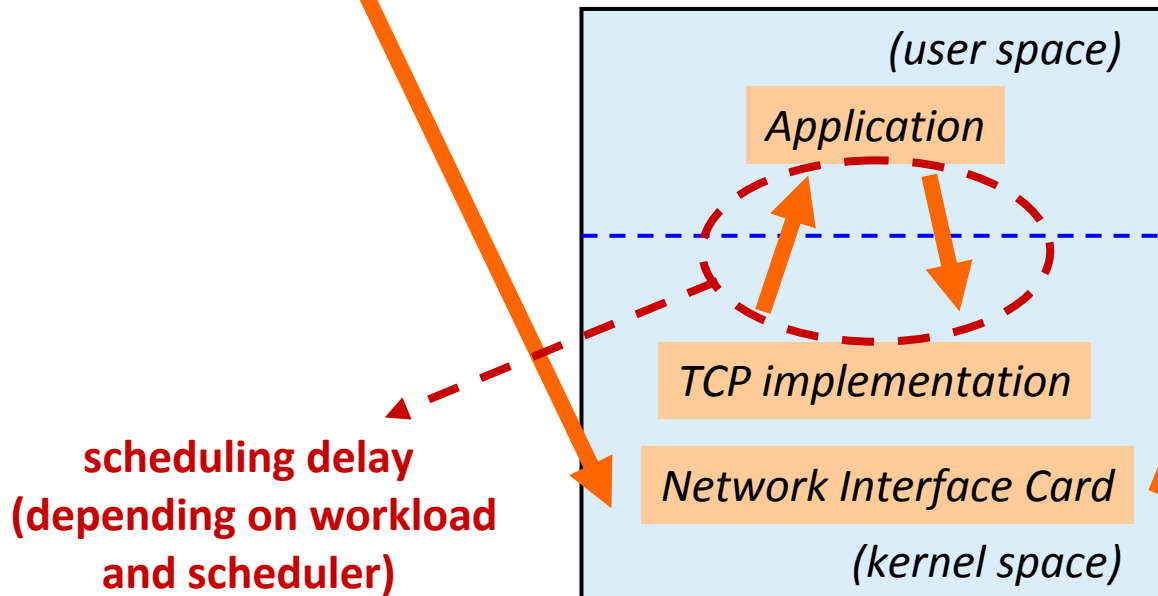
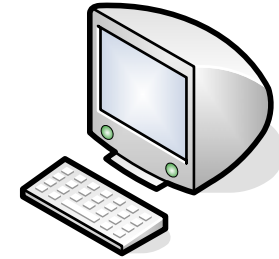
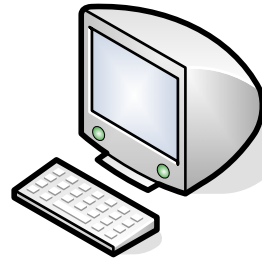
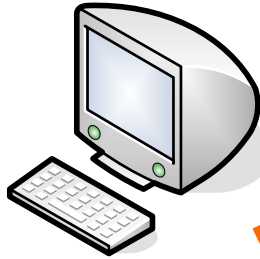




# The Problem



currently relaying is achieved at  
*application-level*



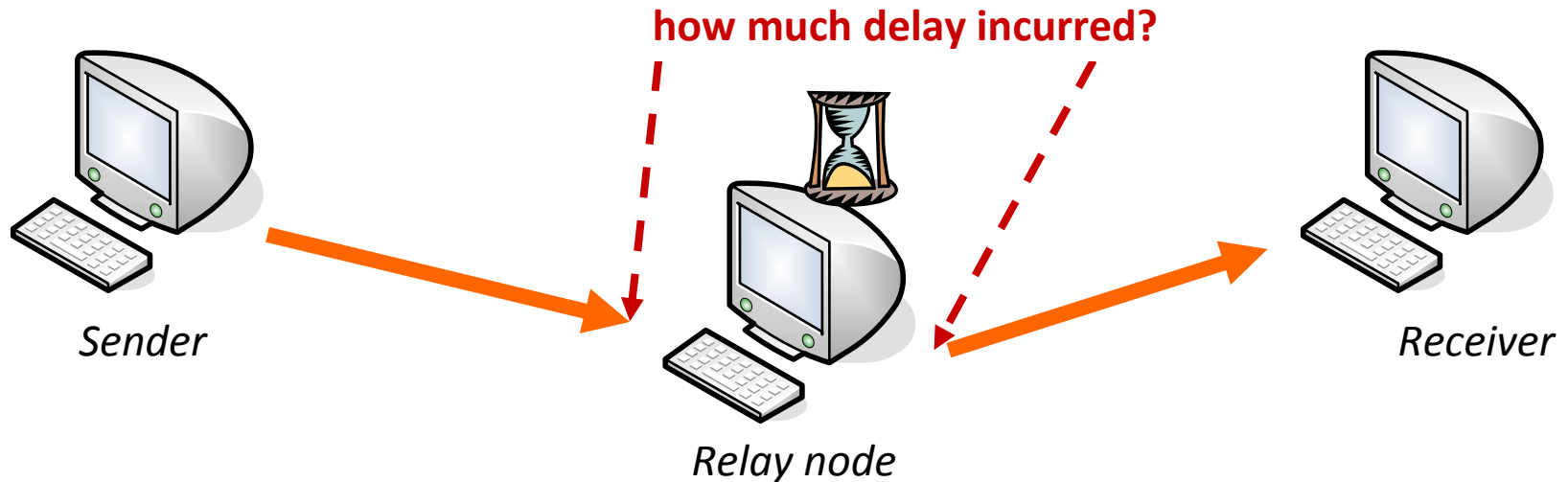
# What We Do

- **Measure** the processing delays at relay nodes  
proposed a measurement methodology and performed a large-scale Internet measurement (> 1000 nodes)
- Understand the **characteristics** of processing delays  
unreliable and predictable
- Examine whether these processing delays **degrade** application quality  
yes, they can significantly affect VoIP quality

# Talk Progress

- Overview
- ➔ ■ Measuring processing delay (PD)
- Large-scale measurement
- PD characterization
- Impact of PD on VoIP quality
- Conclusion

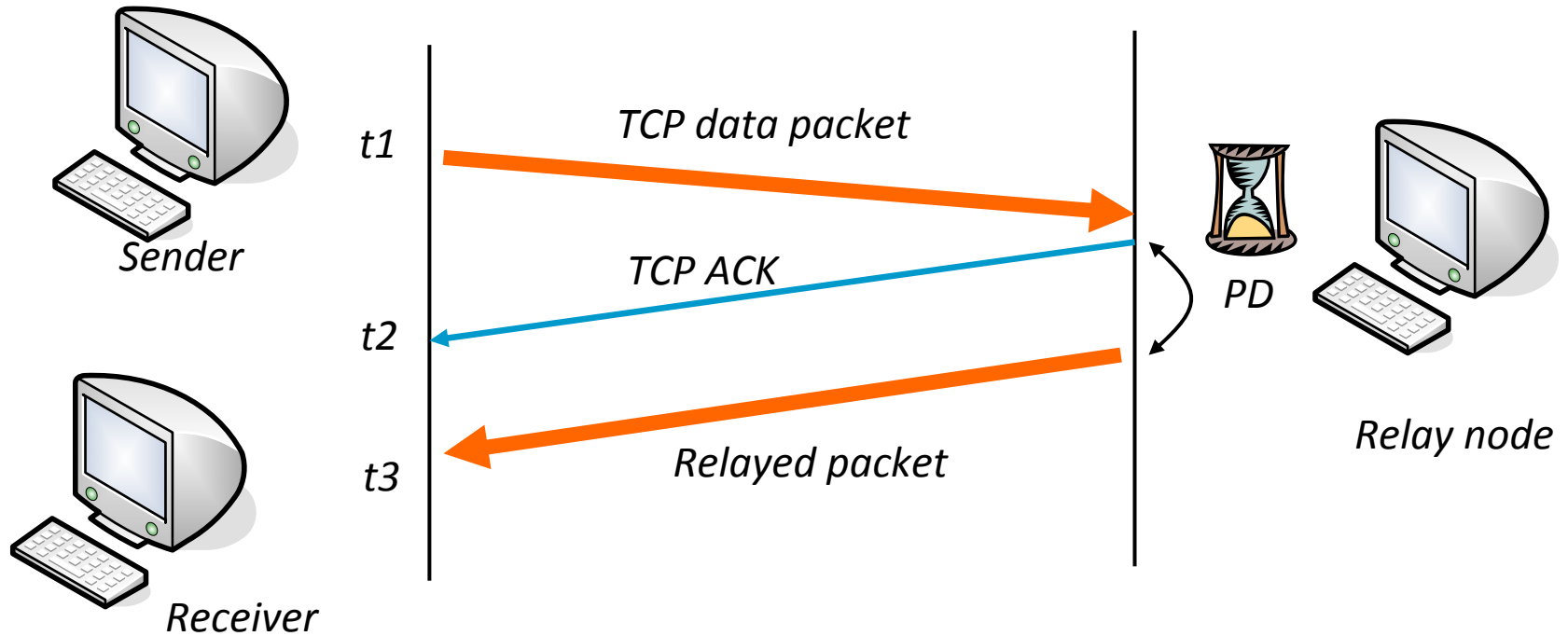
# Our Basic Method for Measuring PD



## Assumptions

- **No intentional delays** at the relay node
- Source transmits packets with **TCP**
- TCP implementation is running at **high priority**, while application is running at low priority
- Sender and receiver are **colocated** at the same place

# The Idea



$$PD = t_3 - t_2$$

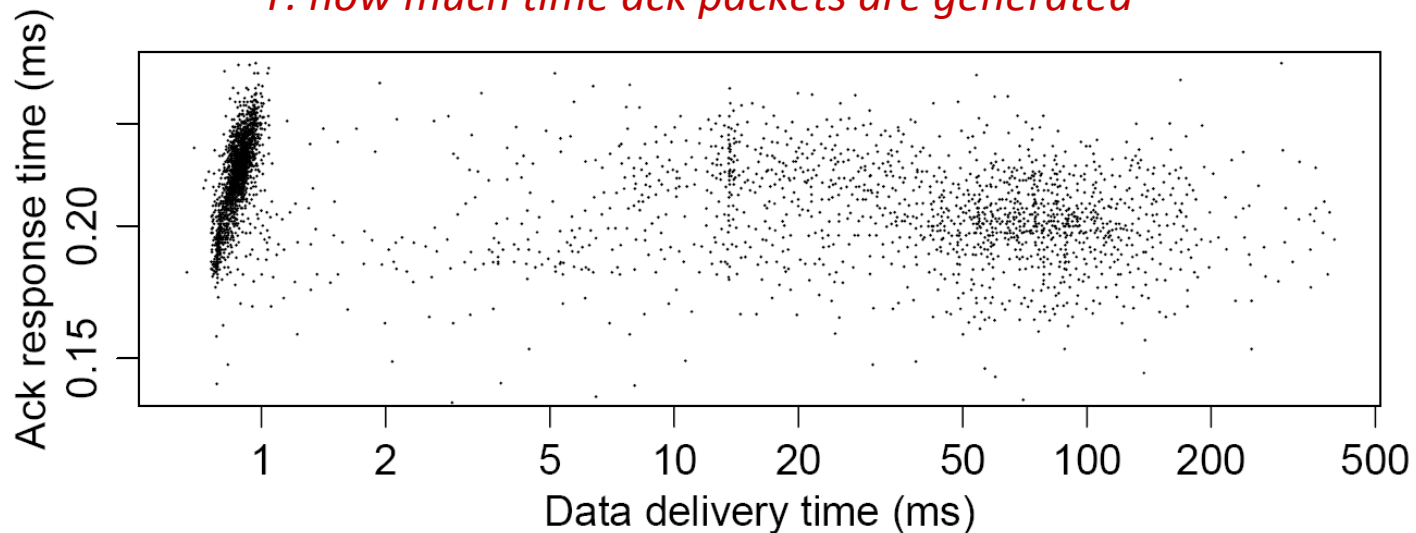
*if network delay is approximately constant*

# Verifying The Assumption:

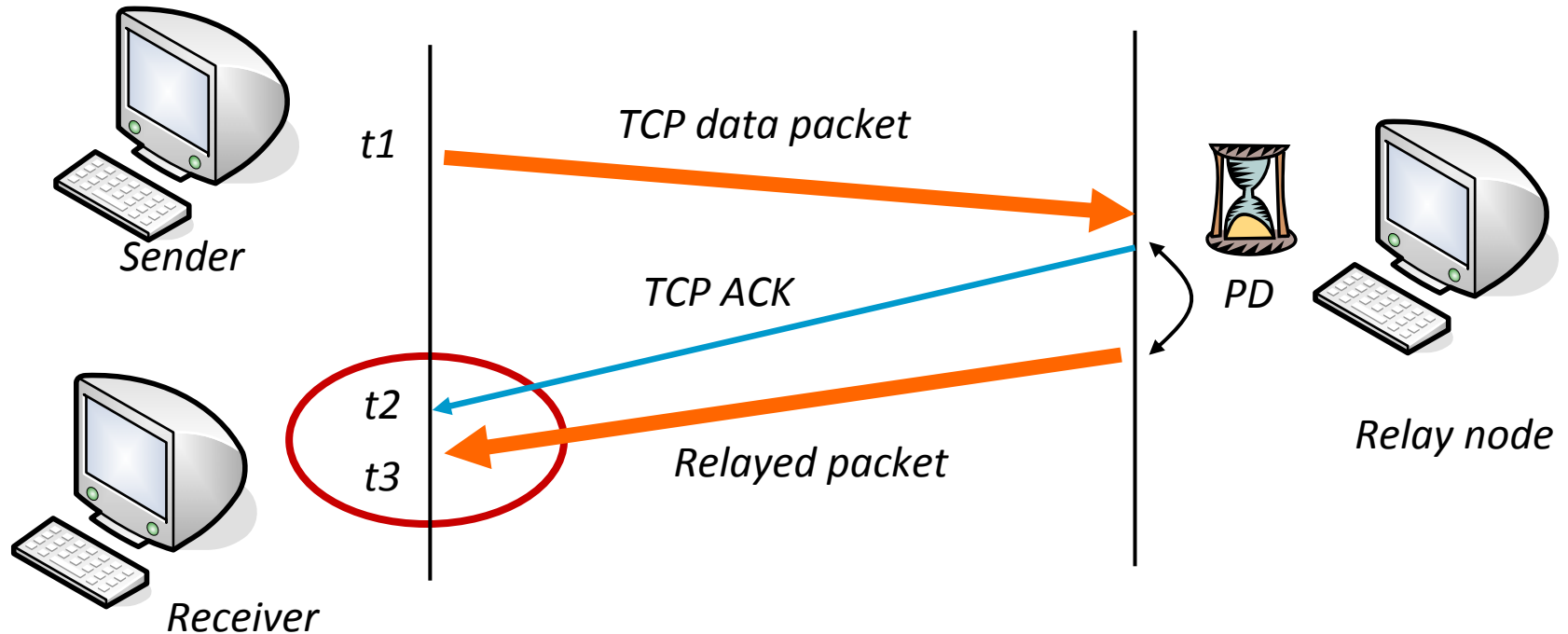
## “Is TCP implementation running at high priority?”

- LAN experiment
- Windows XP on commodity PC
- 200-byte packets sent at 30pkt/sec
- 10 random movie clips to simulate workload

*X: how much time relay packets are generated*  
*Y: how much time ack packets are generated*



# The Problem of Our Basic Method



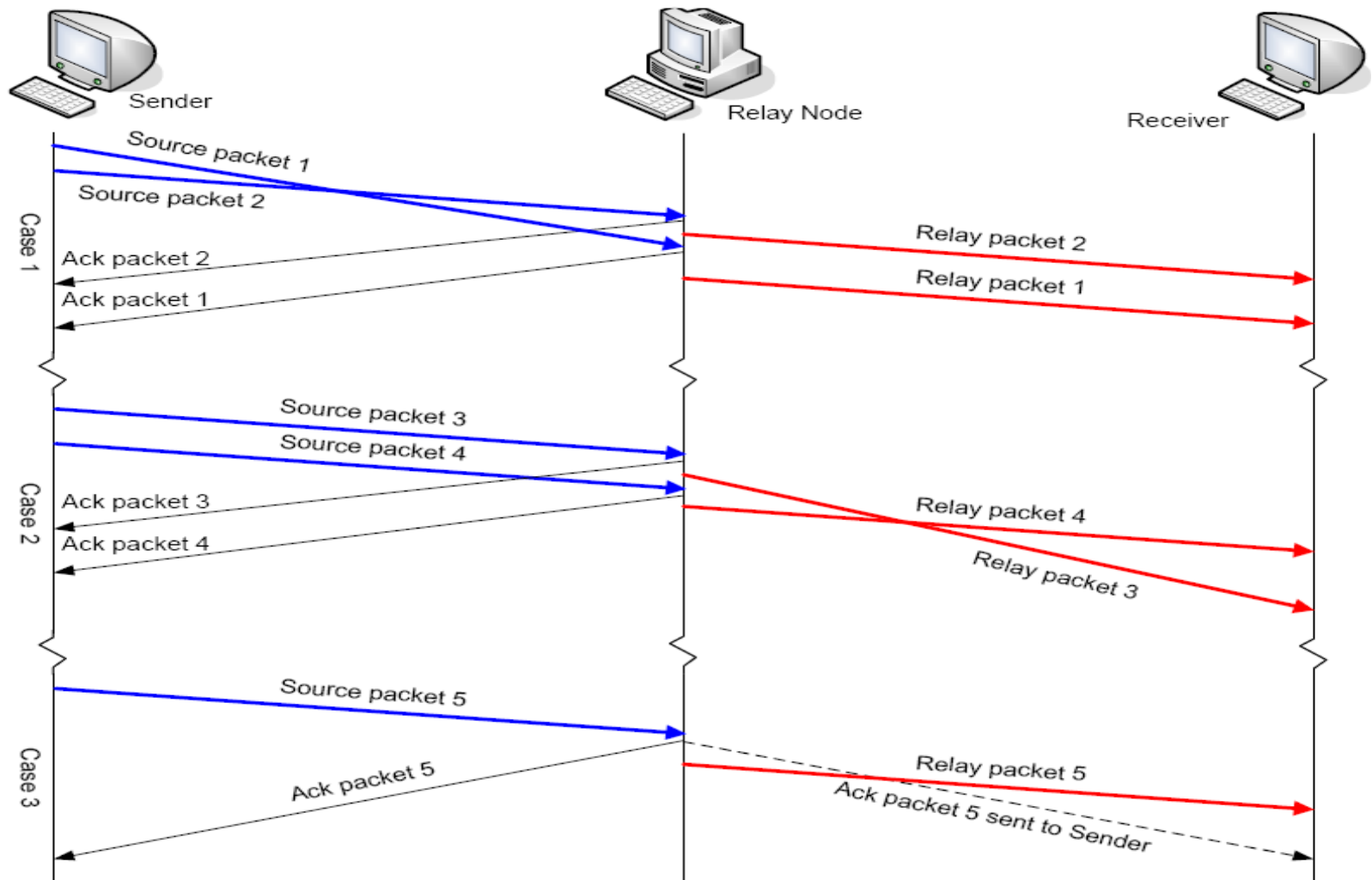
*Network delays are NOT constant  
(mostly because of queueing delays)*

$$\rightarrow t_3 - t_2 \neq PD$$

# Proposed Improvement: IPID Filtering

- Filter out seriously delayed packets
- IPID: the ID field of IP packets
  - **monotonically increasing**
  - infer the order of data packets generated by the sender
  - infer the order of ack packets generated by the relay node
  - infer the order of relay packets generated by the relay node
- If any pair of packets is **reordered** along their transmission
  - ➔ discarding both packets
    - LIS (longest increasing subsequence) algorithm

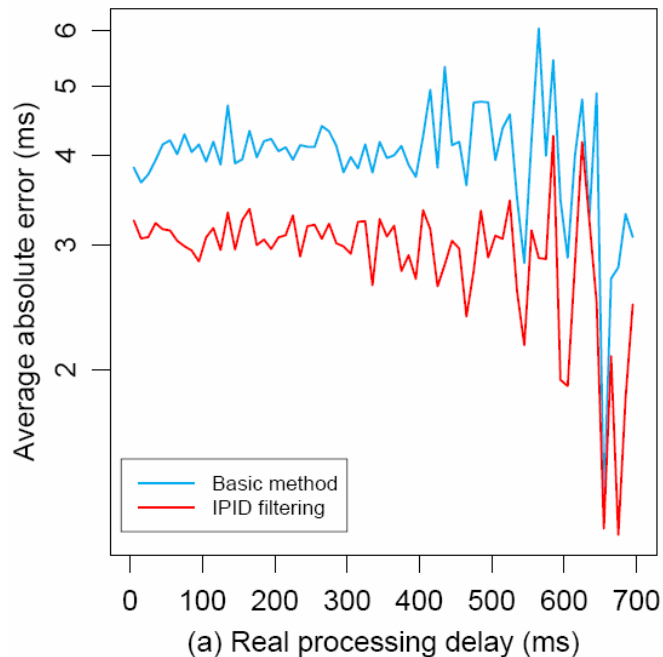
# Some Common Network Reordering Scenarios



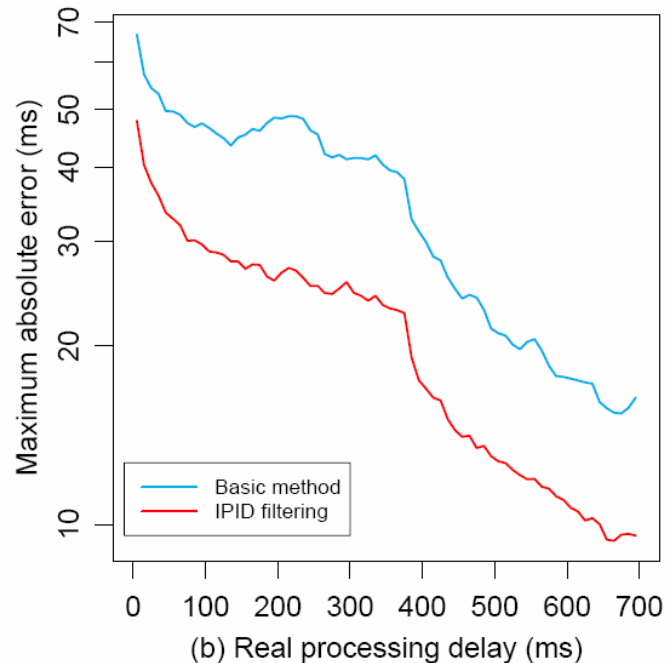
# Effectiveness of IPID Filtering

- LAN experiments
- Evaluate the differences between measured PD and true PD

*Average error of inferred PD*

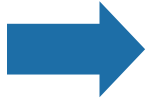


*Maximum error of inferred PD*



# Talk Progress

- Overview
- Measuring processing delay (PD)
- **Large-scale measurement**
- PD characterization
- Impact of PD on VoIP quality
- Conclusion



# Large-Scale Measurement

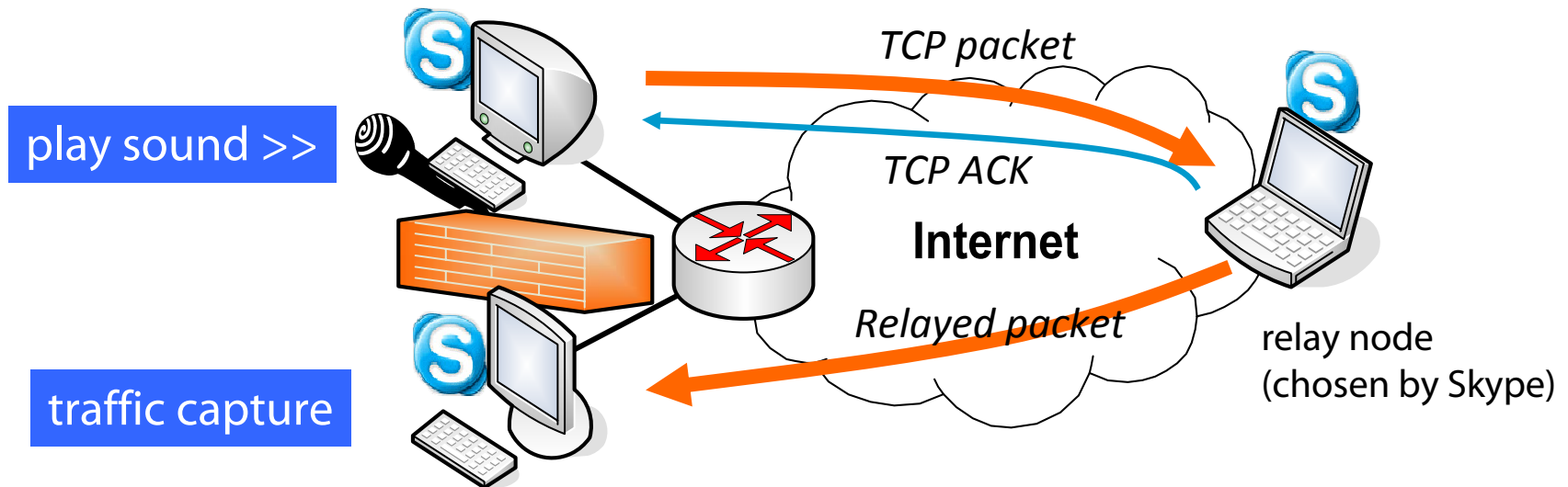
- Goal: To understand PD at real-life relay nodes

- Testbed: 

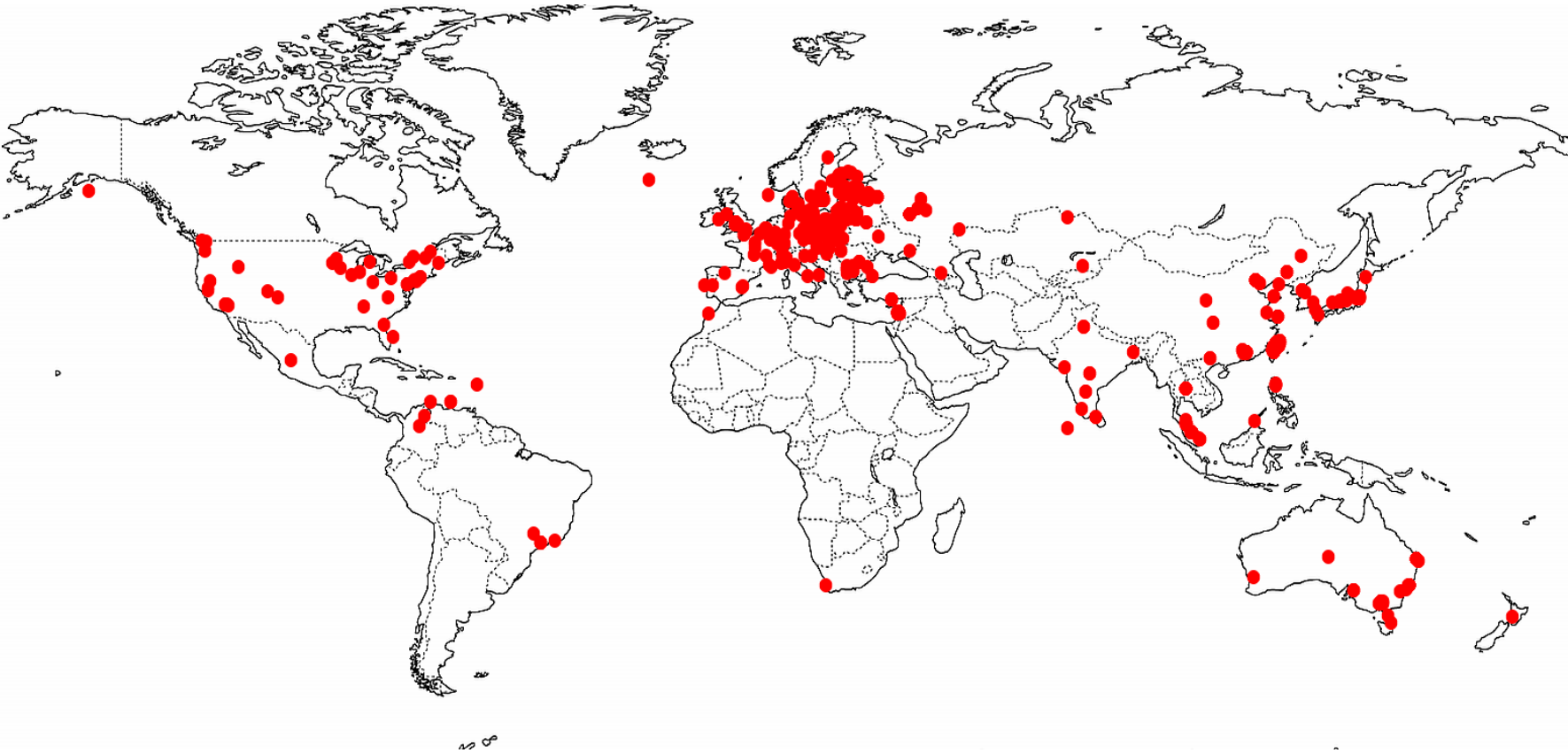
- > 200 thousands super nodes available anytime
- Skype finds a usable relay node automatically if the caller cannot reach the calle
- We can force Skype to pick an alternate super node for relaying a VoIP call

# Collection Methodology

- Automatically call establishment and drop (10 min each)
- Human speech recordings form Open Speech Repository
- Use **firewall** to block caller <-> callee direct connection
- We block used relay nodes to force Skype find a new one



# Geographical Diversity of Observed Relay Nodes



## Call Information

# Calls	Time	# Pkts/call	ART	DDT
1,115	10 min	19,210 pkts	203 ms	212 ms


*On average, data packets incur 9 ms PD*

## Estimated Processing Delays

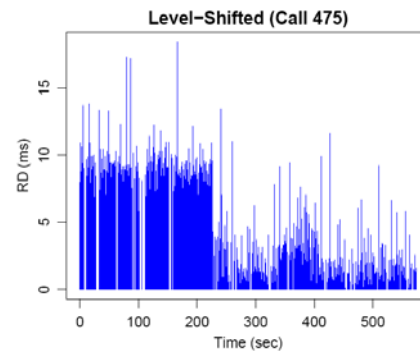
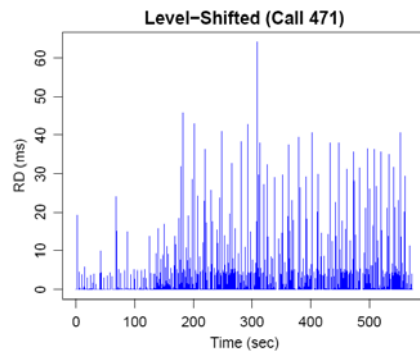
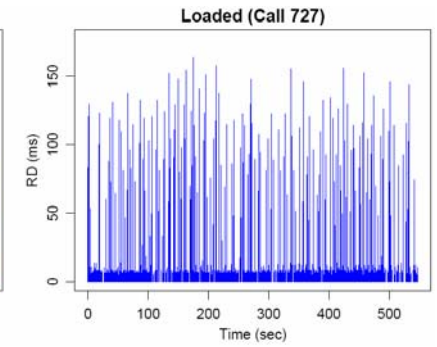
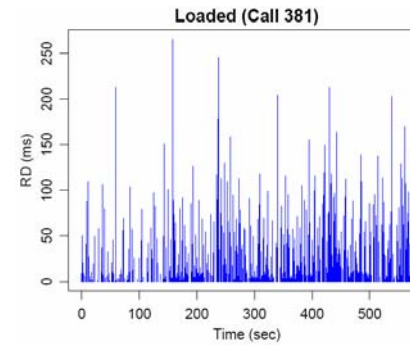
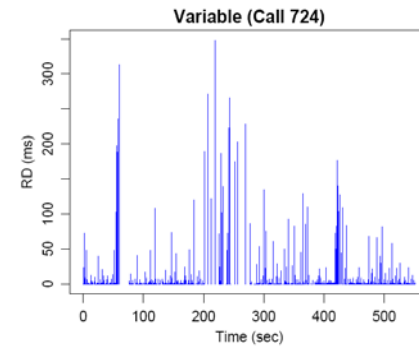
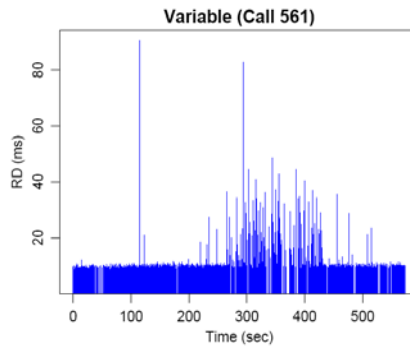
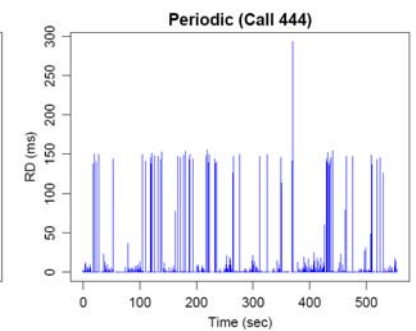
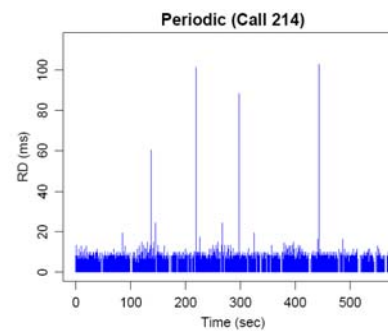
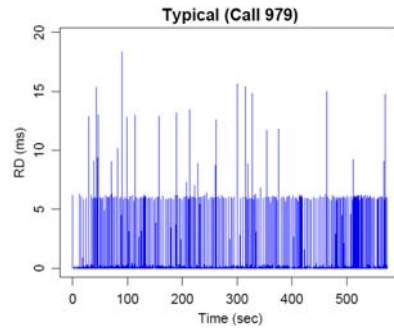
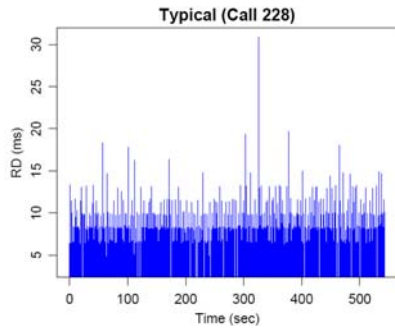
# Samp.	Samp. dens.	PD (Avg / Max / SD)
15,739 per call	26 samp. / sec	5 ms / 239 ms / 7 ms

*On average, maximum PD is around 240 ms*

# Talk Progress

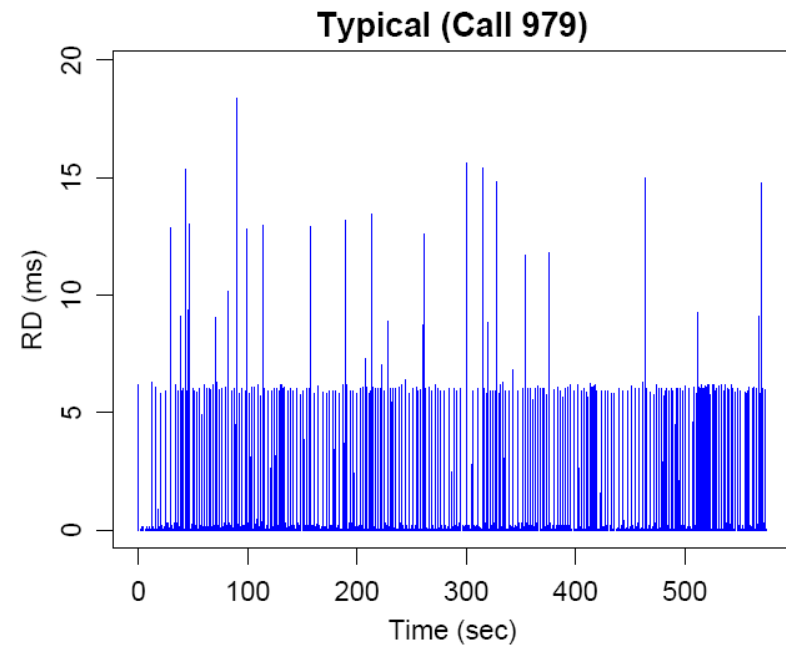
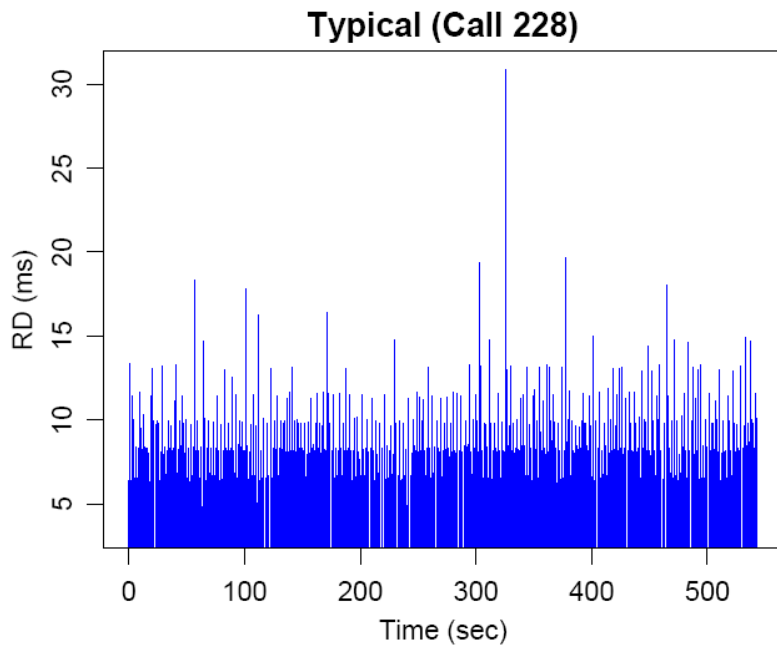
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# A Classification of PD Patterns



# A Classification of PD Patterns (1)

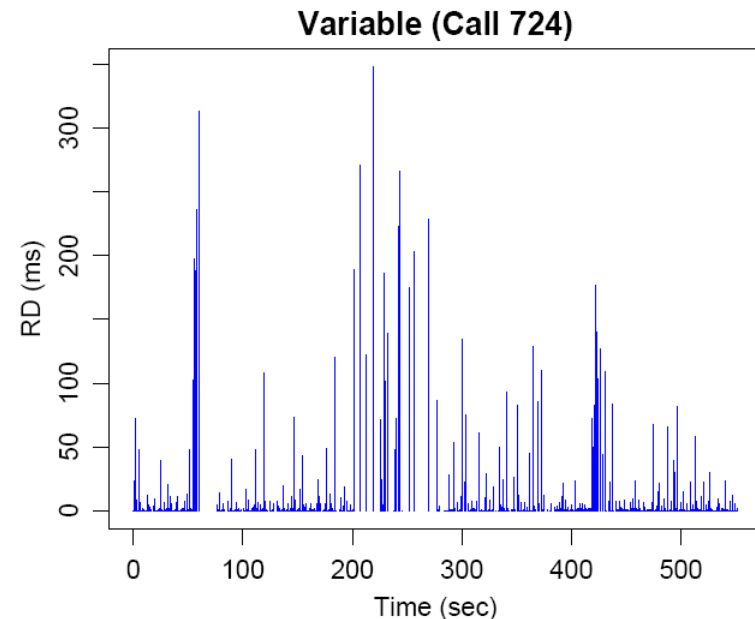
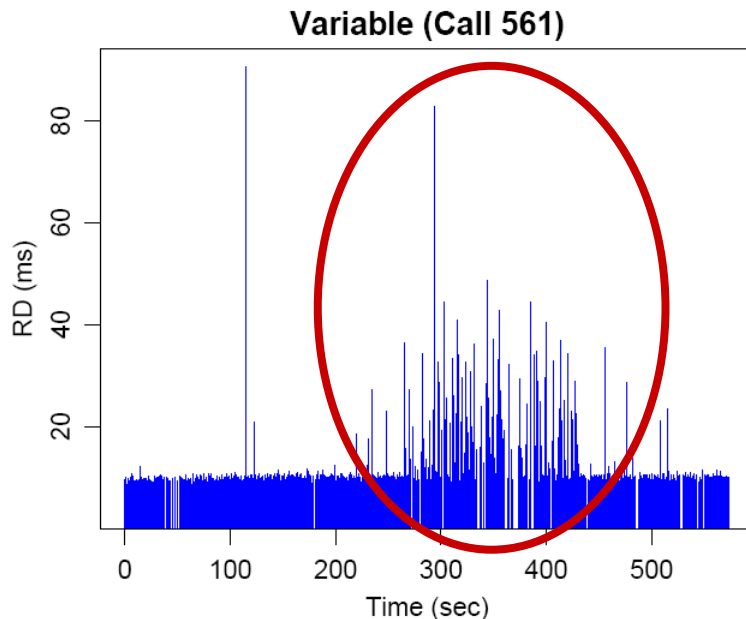
## *Typical Processing Delays*



- *variation of PDs is small*
- *magnitude is rarely higher than 20 ms*
- ➔ *the relay node is lightly-loaded and the computer is not in use*

# A Classification of PD Patterns (2)

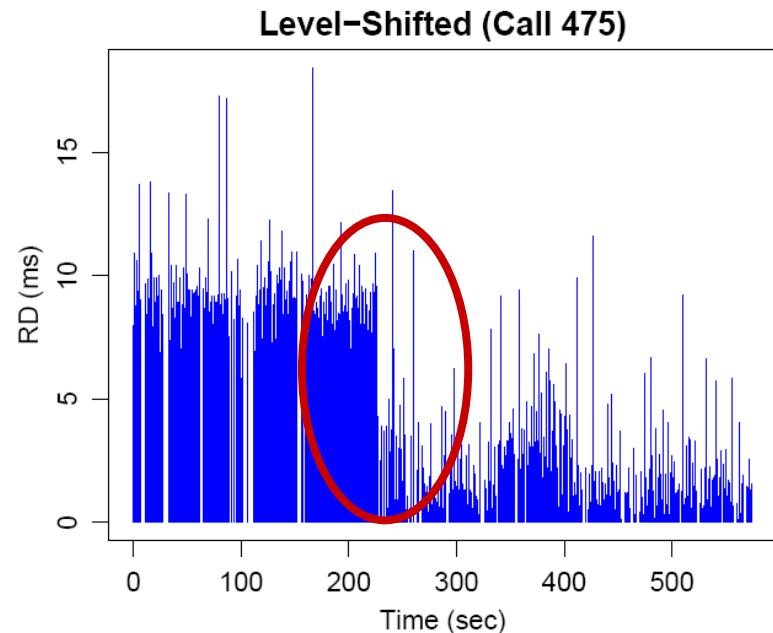
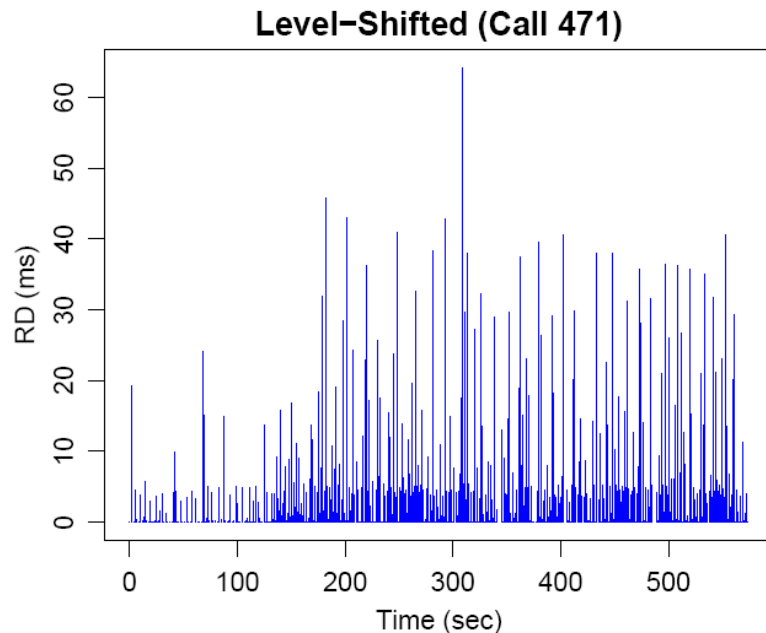
## *Variable Processing Delays*



- *PDs are stable most of time, but occasionally very different*
- *e.g., the PDs of call 561 are relatively high during the 250–450 ms period, which is likely due to a person using the computer, or an application with a time-varying workload is running.*

# A Classification of PD Patterns (3)

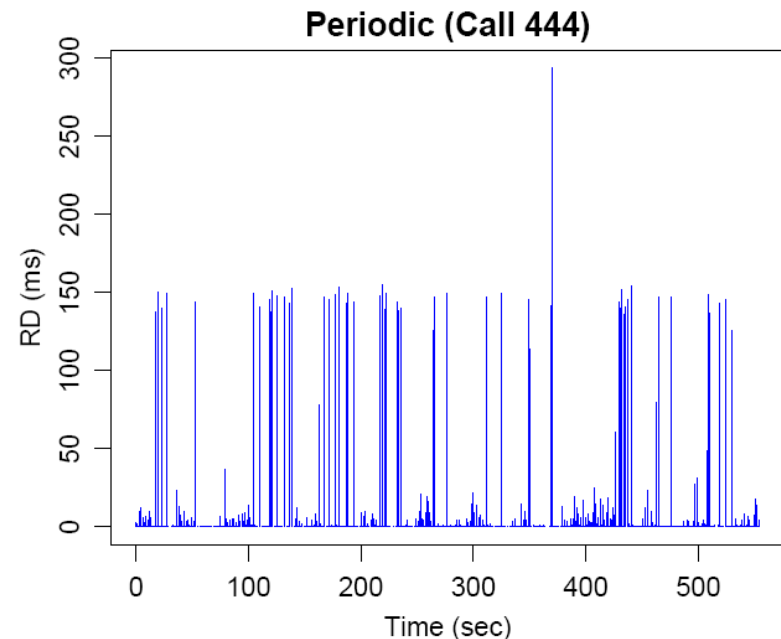
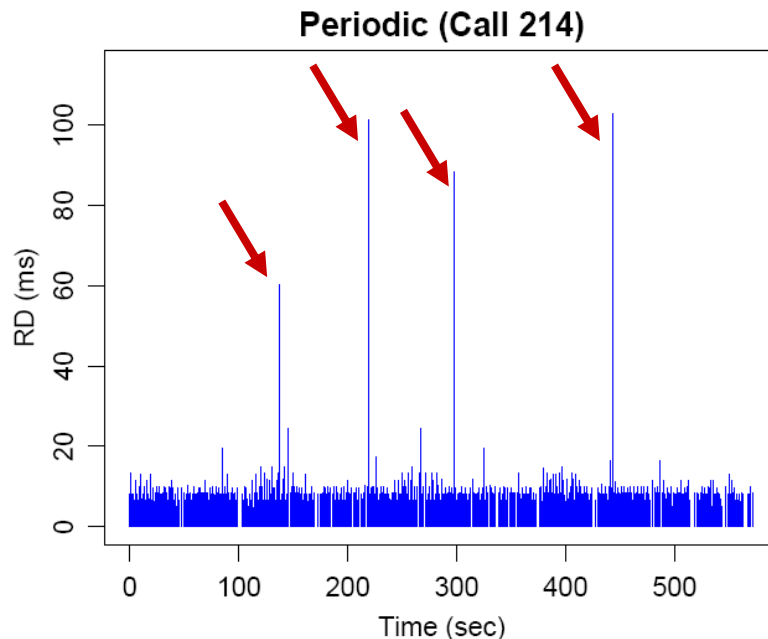
## *Level-shifted Processing Delays*



- *The levels of PDs are increased or decreased by a significant magnitude, say, larger than 10 ms*
- ➔ *likely a heavily loaded task starts or stops running on the relay node*

# A Classification of PD Patterns (4)

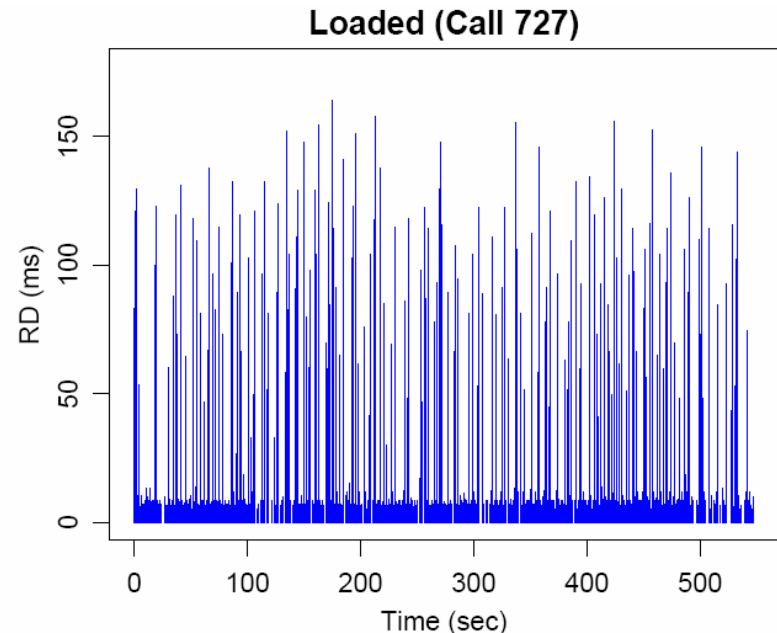
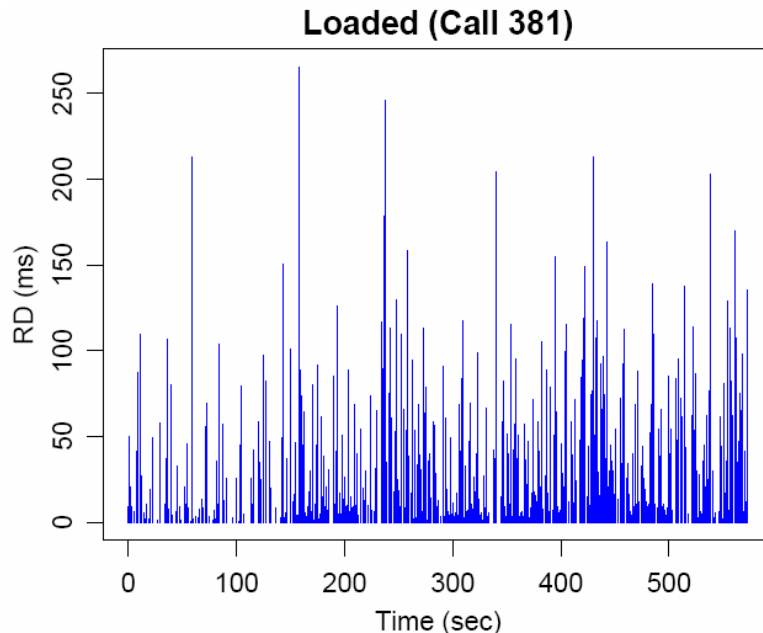
## *Periodic Processing Delays*



- *Bursts of high PDs occur at regular intervals, possibly because of the behavior of an application.*
- *e.g., the 1-minute interval in Call 214 may be caused by an email notification program with a one-minute check interval.*

# A Classification of PD Patterns (5)

## *Loaded Processing Delays*

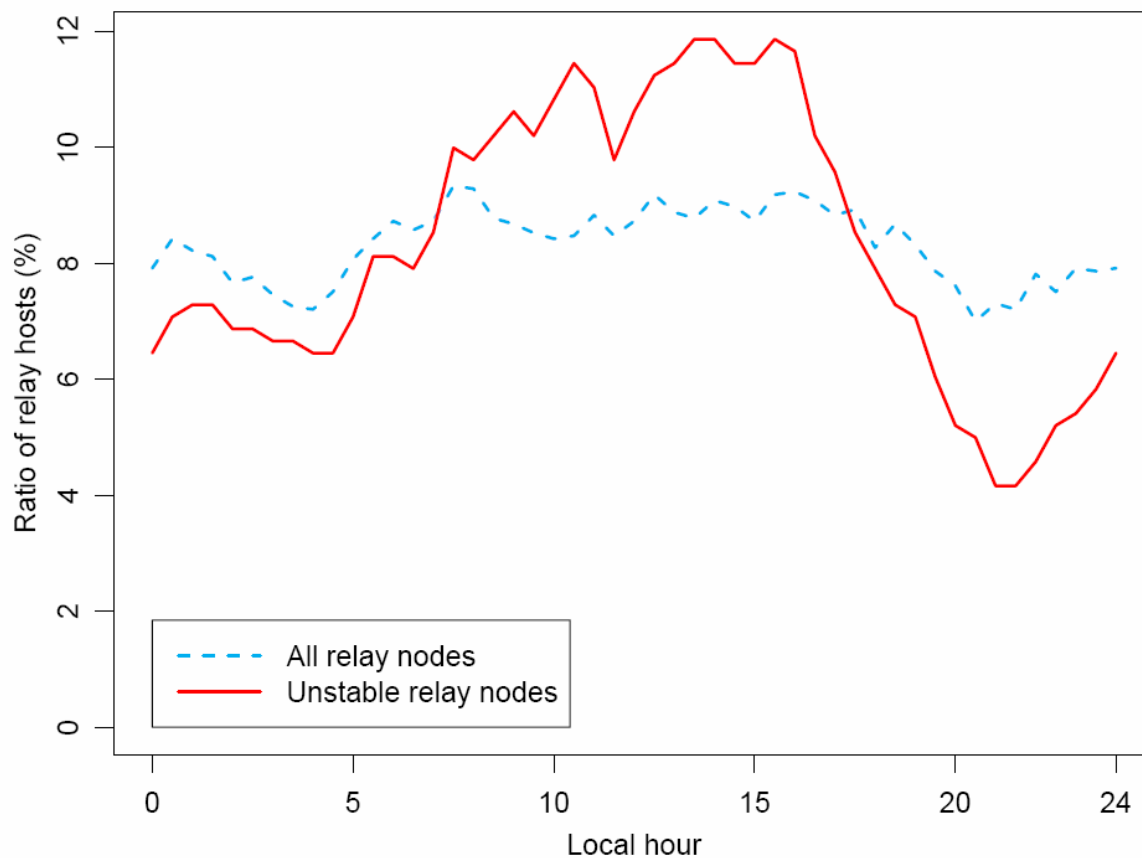


- *The level of PDs remains large, say 100 ms or higher.*
- ➔ *the relay node should be running computation- or I/O- intensive applications*

# Busy Levels


- The level of workload on the relay node
- 95% percentile of PD within a 10-second window
  - Lightly-load host: application always serviced when needed
  - Even on a heavily load machine, the PD incurs by relay packets are not always large
    - A packet arrives just before the application's execution quantum → get serviced immediately
    - otherwise → has to wait some time
- If busy level changes  $> 10$  ms in successive 10-second windows → unstable relay nodes

# Average Ratios of All Relay Nodes and Unstable Relay nodes



*Unstable nodes are mostly observed in work hours at local time*

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# Impact of PD on VoIP Quality

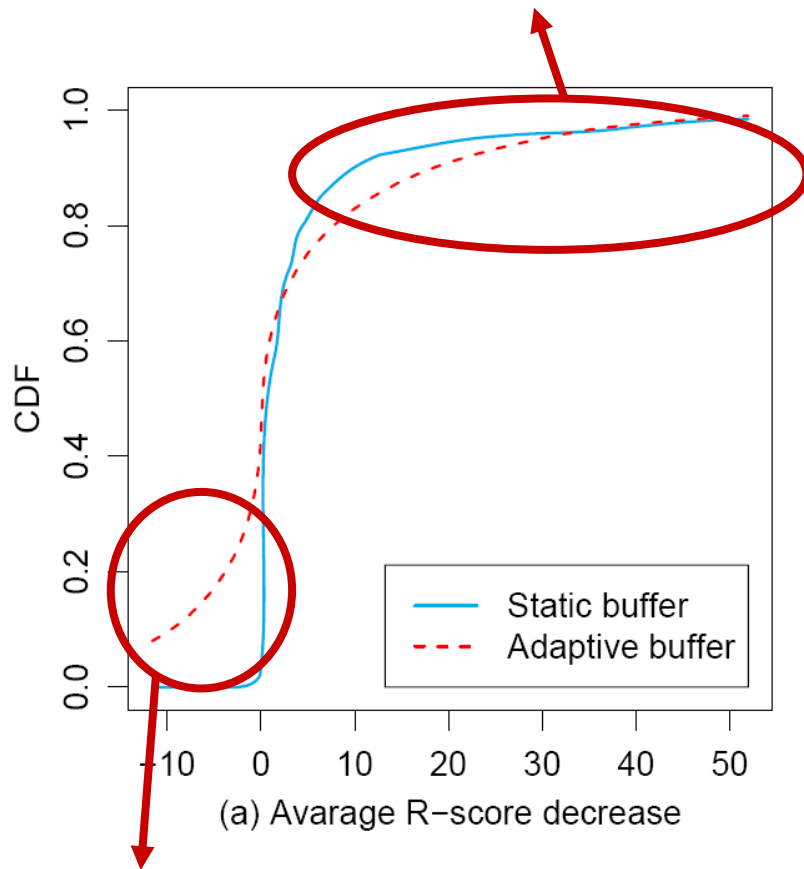
- p2p relaying is commonly used by VoIP applications
  - bypass firewall/NAT
  - network latency may be reduced by overlay routing
  - VoIP bandwidth requirement is not high, e.g., 32 Kbps
- Use trace-driven simulations to assess the impact of PD
  - combine different network delays and processing delays
  - simulate VoIP calls with and without processing delays
  - examine their differences in VoIP quality

# Methodology

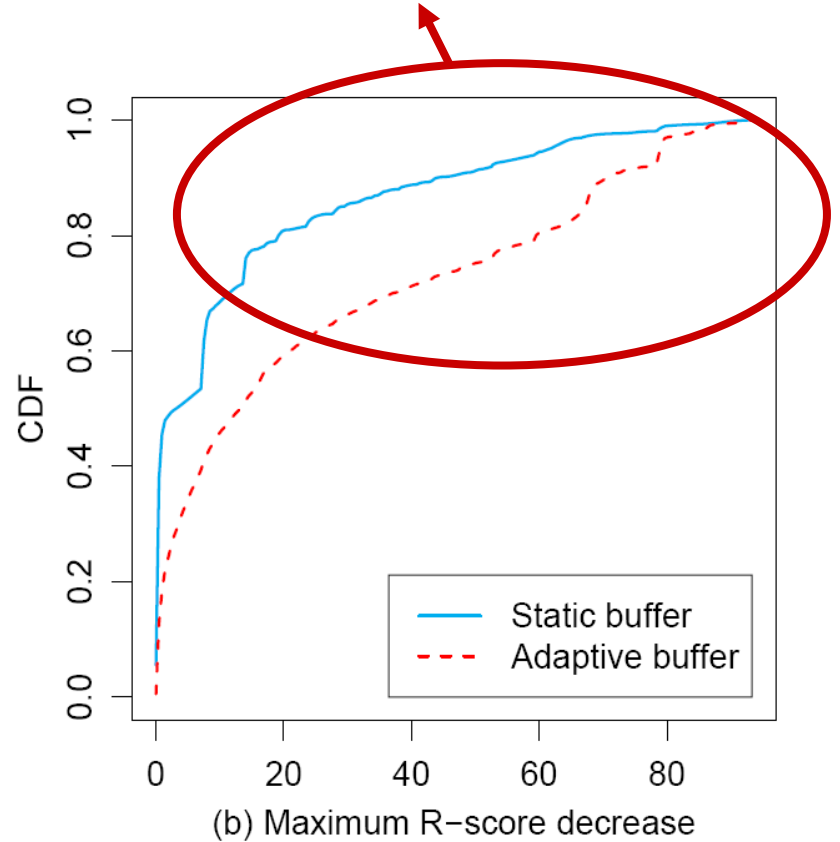
- Measuring VoIP quality degradation
  - ITU-T E-model (G.711) based on delay and loss rate
  - yields a R-score ranges from 0 -- 100
  - R-score > 80 → satisfactory
  - R-score < 70 → unacceptable
- Two classes of VoIP playout buffer sizing schemes
  - **static buffer** (60 ms according to Skype)
  - **adaptive buffer** (according to Ramjee INFOCOM'94)
    - size = mean delay + 4 x standard deviation of delays
    - mean and std dev are derived from EWMA processes
- 50,000 random samples from 1,243,225 (1,115 x 1,115) possible configurations

# R-scores Decrease due to PD

*20% of calls incur significant average R-score decrease*

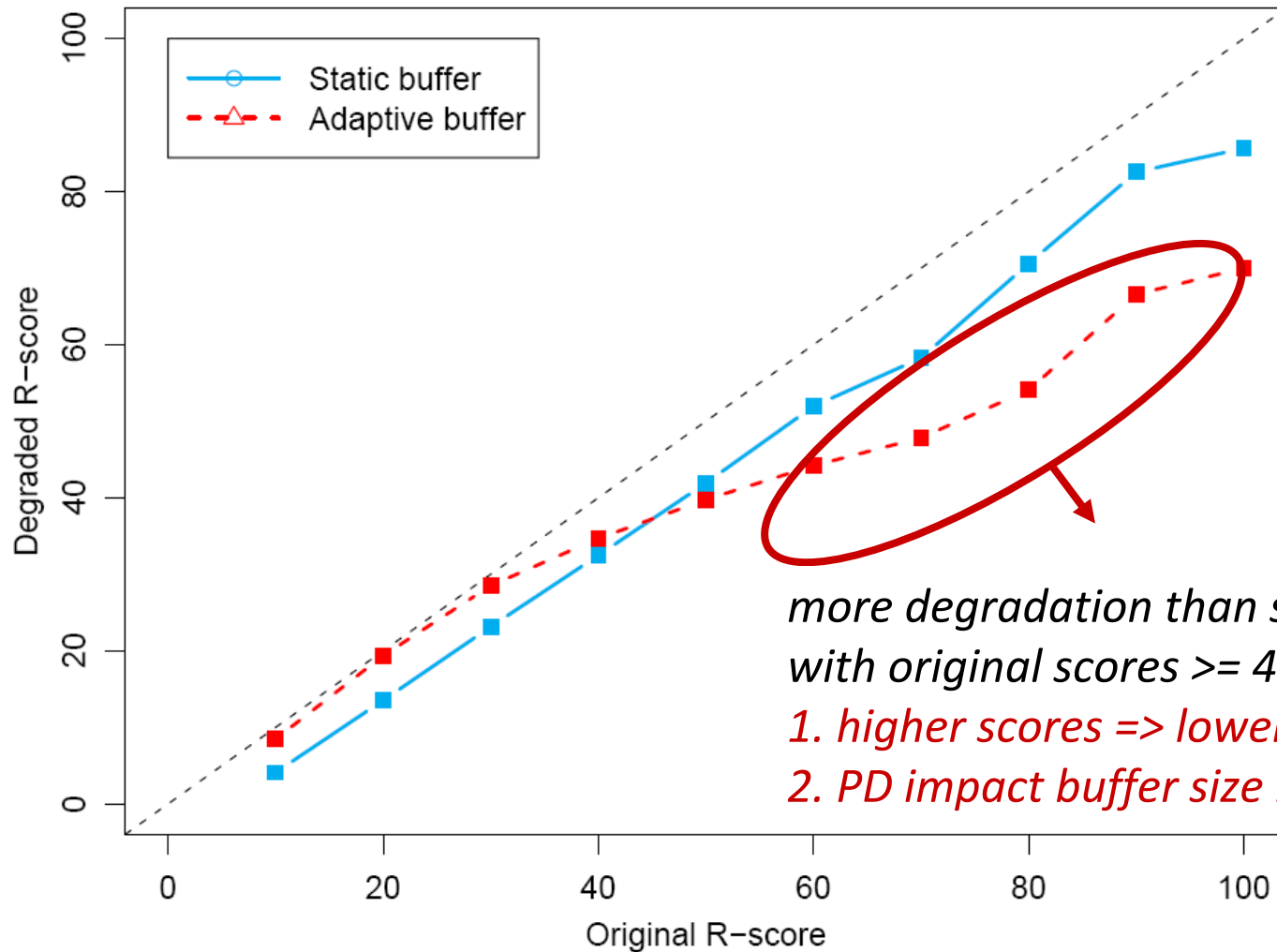


*40% and 60% of calls incur significant maximum R-score decrease*



*PD leads to larger playout buffer that leads to lower e2e packet loss rate*

# Original R-scores vs. Average R-scores

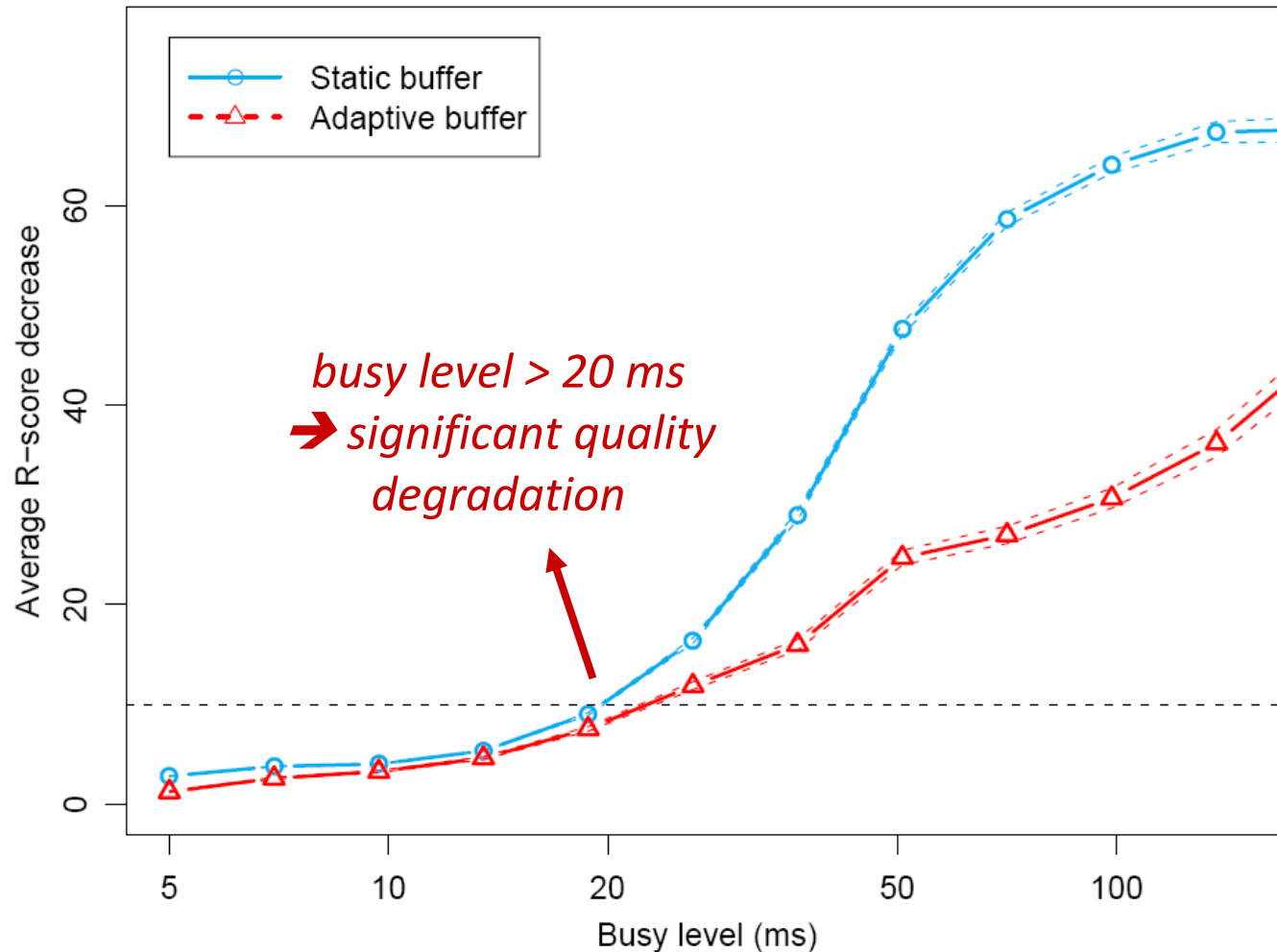


*more degradation than static buffer  
with original scores  $\geq 40$*

- 1. higher scores => lower net delay*
- 2. PD impact buffer size more*

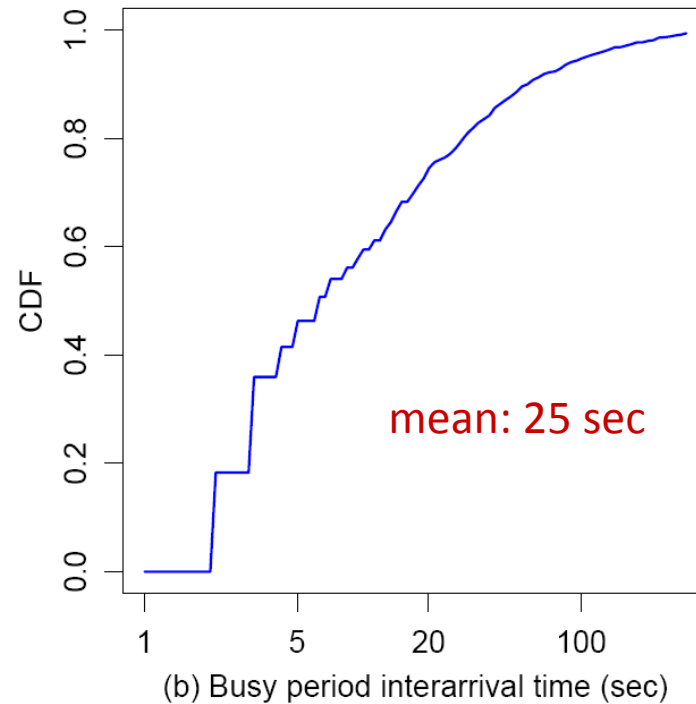
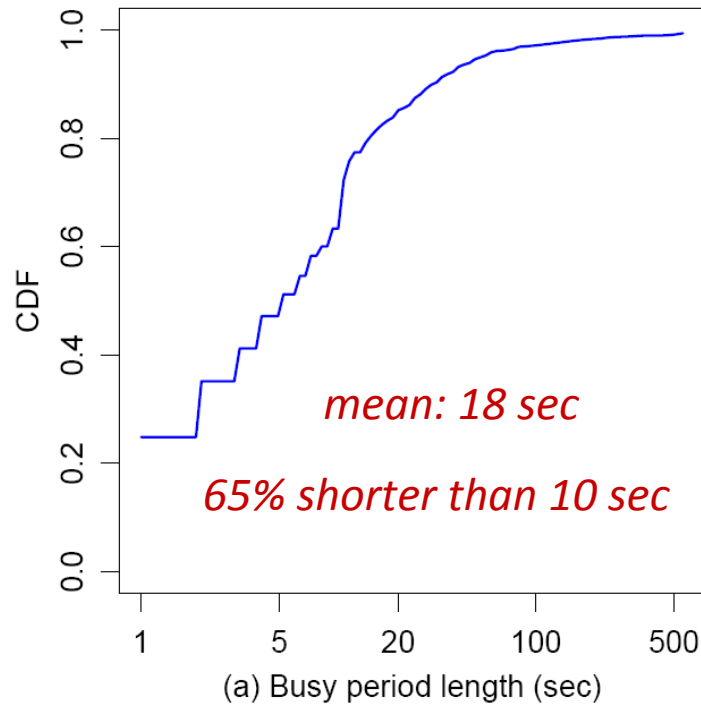
# Impact of Busy Levels

What busy levels of PD would significantly impact VoIP quality?



# Busy Period Characterization

- We define a node as busy when its busy level is higher than 20 ms
- 23% of relay nodes were ever busy during a 10-minute call



The nodes tend to switch between busy and non-busy states frequently

# Summary

- consider a hidden aspect of peer-to-peer relaying --- processing delays at relay nodes
- PD can be **determental** to VoIP quality
- we should avoid a relay node with a **busy level higher than 20 ms**
- we have to monitor the processing delays of a relay node **continuously** (like we do for the network delays)

# Questions?



# Thank You!

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