

# RPT: Re-architecting Loss Protection for Content-Aware Networks

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# Motivation:

## Delay-sensitive communication



### Real-time streams:

FaceTime, Skype, on-line games.

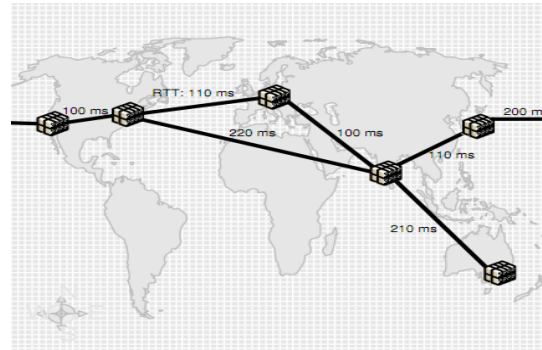
Maximum one way latency  $\sim 150\text{ms}$



### Soft-realtime intra-data center communication

[DCTCP, D<sup>3</sup>]

Response time  
 $\sim 250\text{ms}$



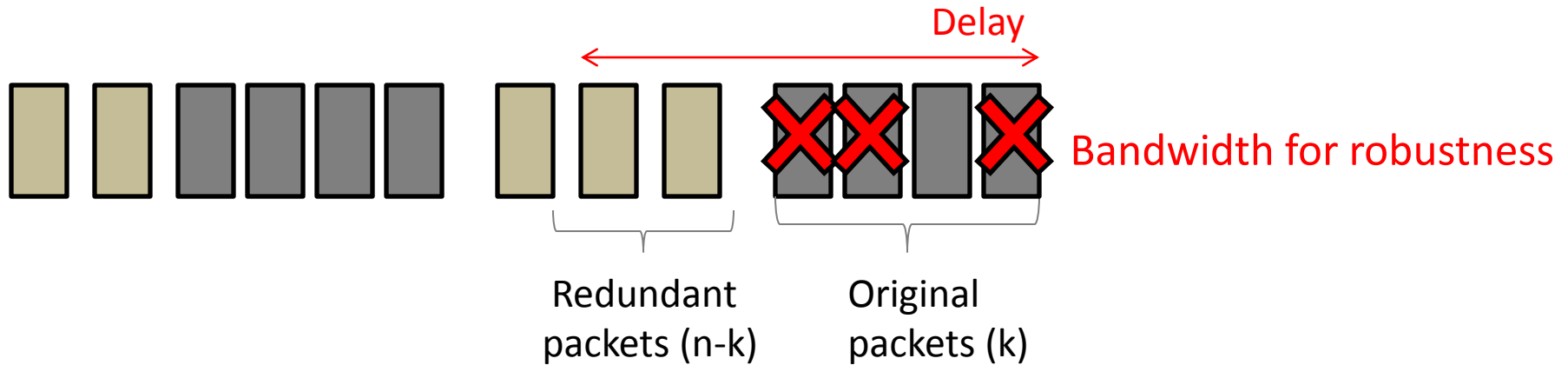
### Time critical inter- data center communication

[Maelstrom]

Minimizing data loss in time-critical communication is important, but challenging because of the time constraint.

# Loss protection today: Redundancy-based recovery

## Forward Error Correction



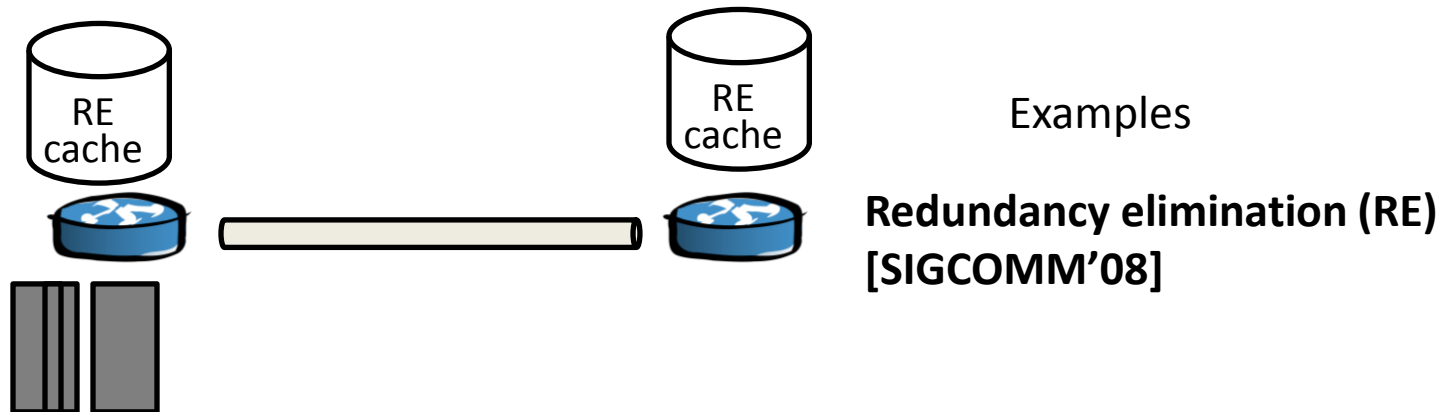
- FEC couples delay with redundancy
- Small batch size makes FEC more susceptible to bursty loss
- Difficult to tune parameters ( $n$  and  $k$ ) [TIP2001,INFOCOM2010]

Amount of redundancy 20%~50% in Skype video[Multimedia'09]

# Content-aware networks changes the trade-off of redundancy

Content-aware networks = caching + content-aware processing to remove duplicates

Caching effectively minimizes the bandwidth cost of redundancy

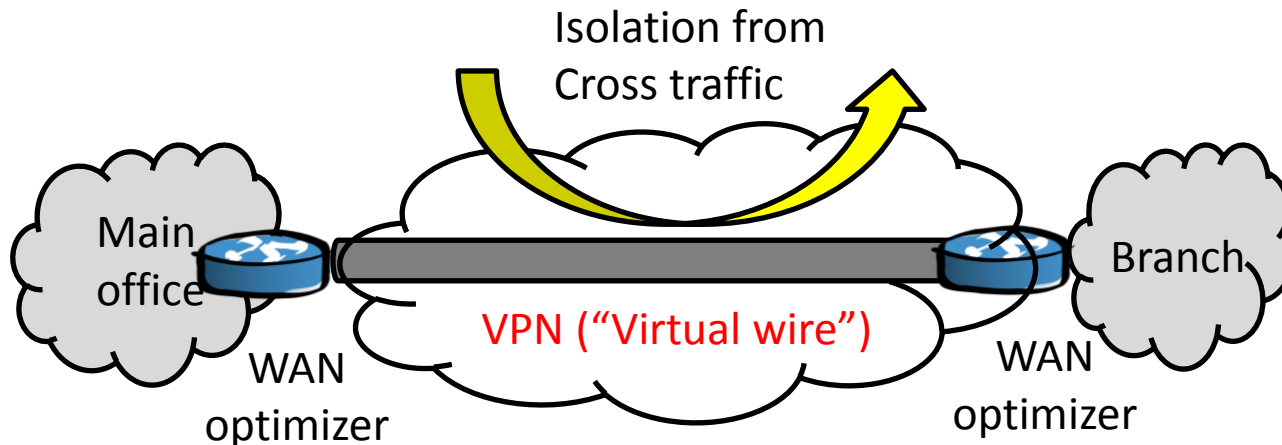


Bandwidth overhead of 100% redundancy: 3%

Content-Centric Networking (CCN) [CoNEXT'09]

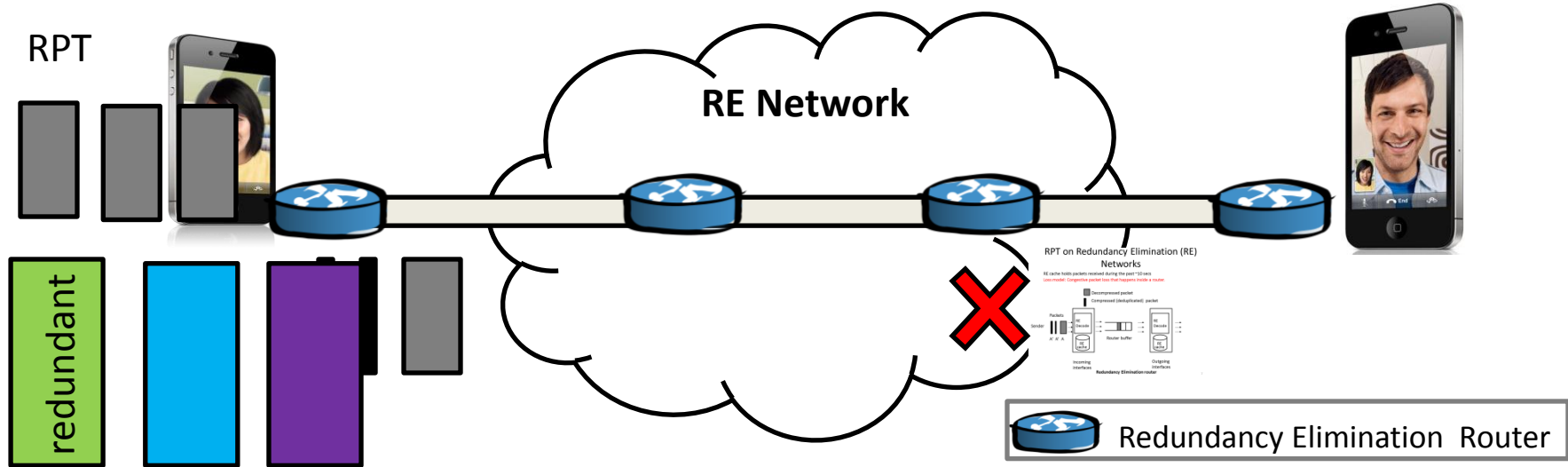
# Deployment of content-aware networks

- Product: WAN optimizers (10+ vendors)
  - Cisco, Riverbed, Juniper, Blue Coat Systems
  - E.g., Cisco deployed RE on 200+ remote offices.
  - Corporate networks
    - Riverbed: 50+ corporate customers, datacenter deployments



# Redundant Packet Transmission (RPT)

- Introduce redundancy in a way that the network understands



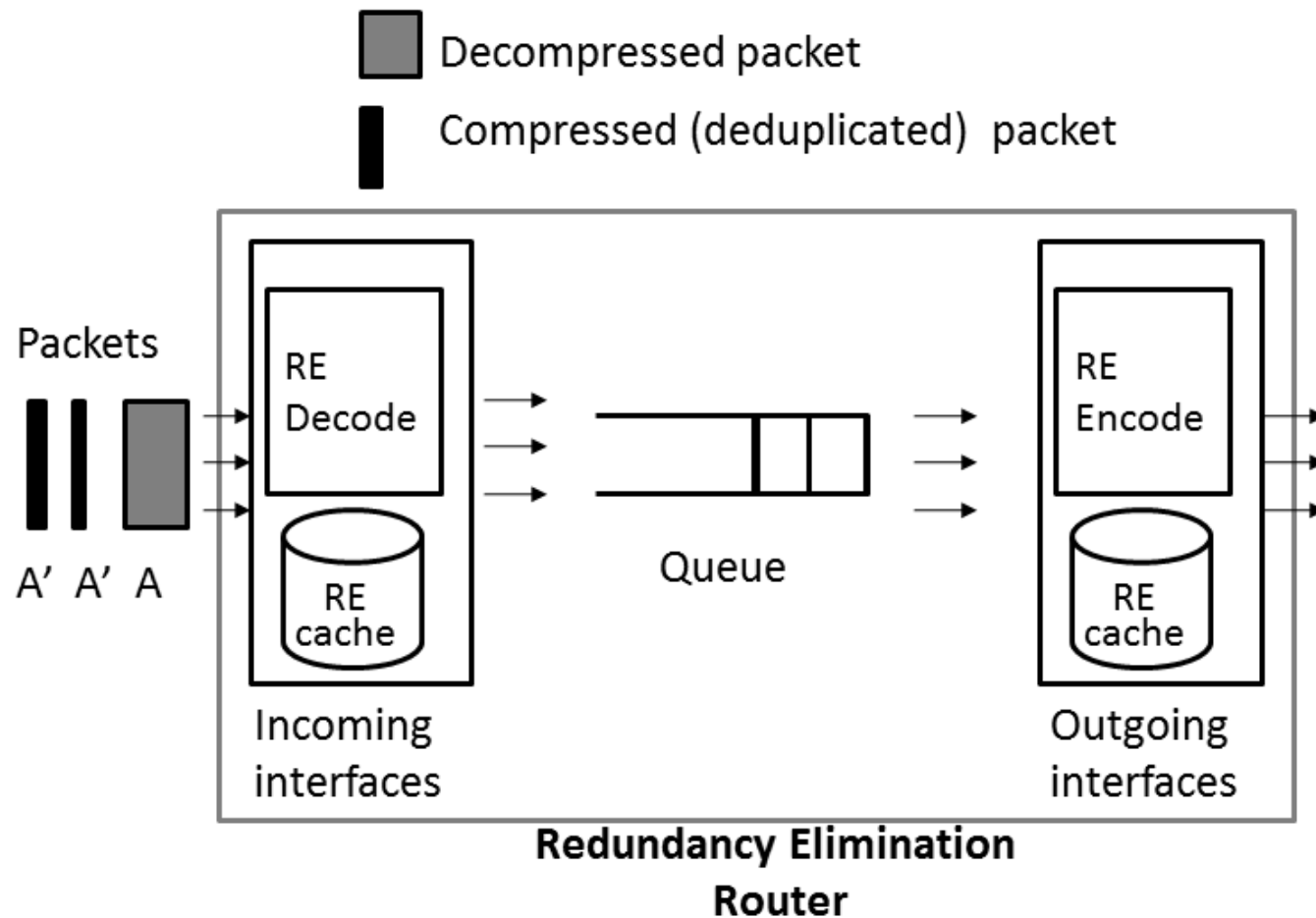
## FEC Questions/Challenges

- How do we make sure we retain the robustness benefits?
- How much redundancy is needed? How does it compare with FEC?
- Is this safe to use?

# RPT on Redundancy Elimination (RE) Networks

RE cache holds packets received during the past ~10 secs

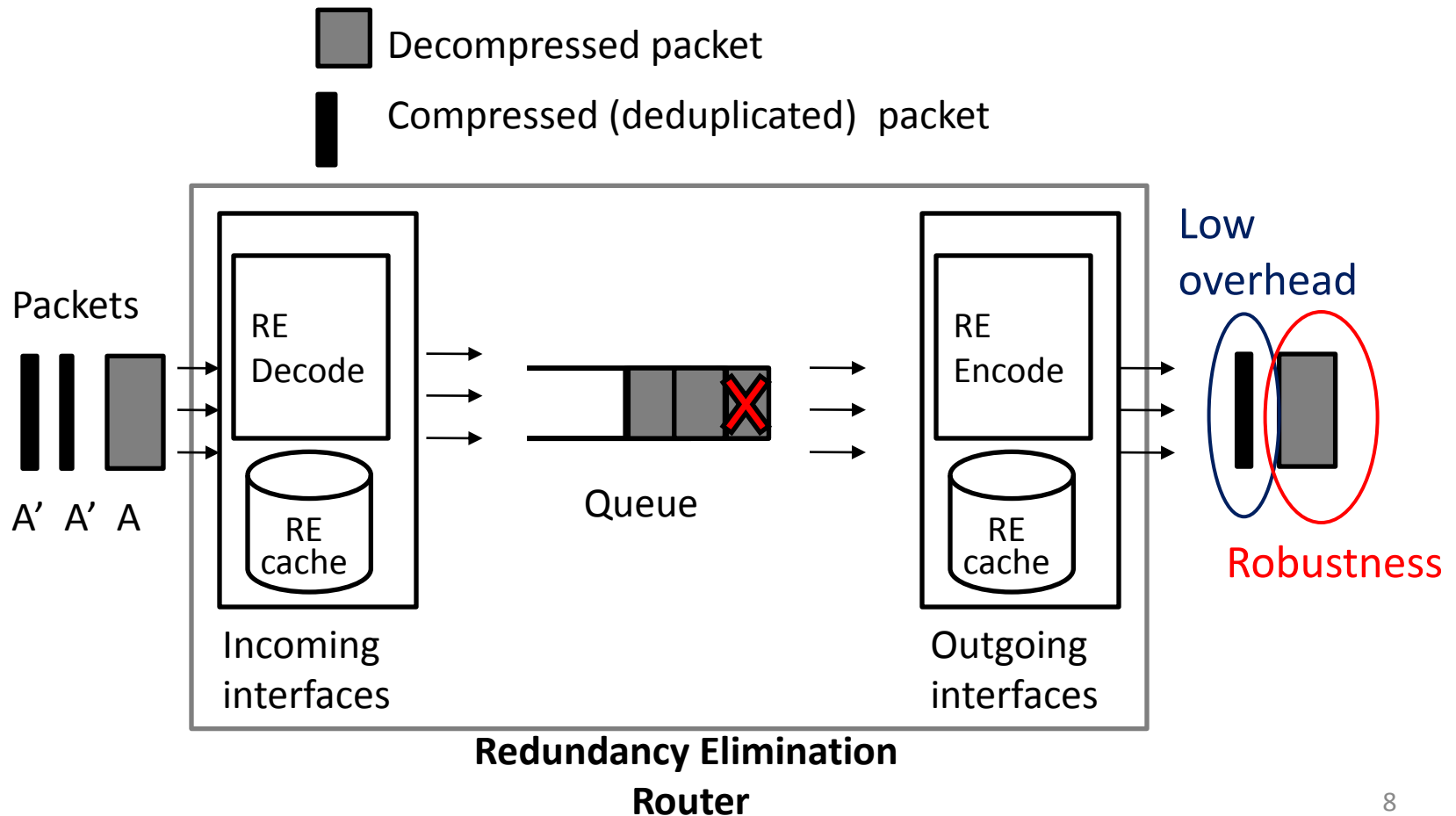
Loss model: Congestive packet loss that happens inside a router.



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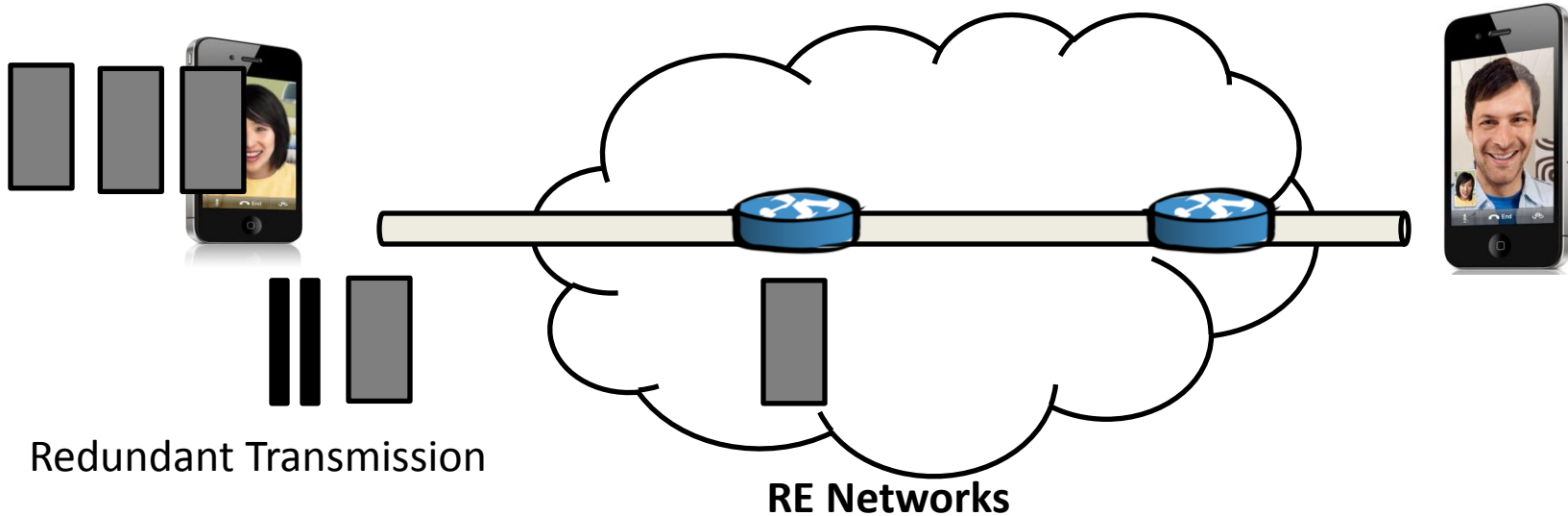
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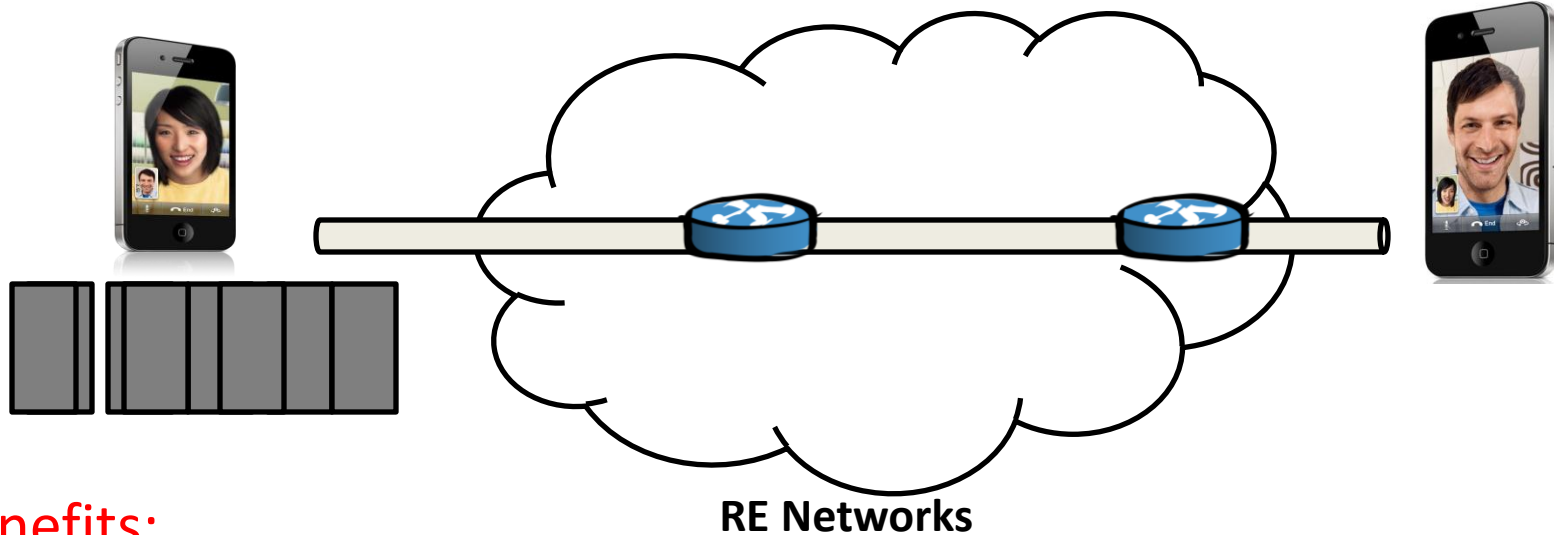
# Redundant Packet Transmission

- Introduce redundancy in a way that the network understands



# Redundant Packet Transmission

- Introduce redundancy in a way that the network understands



## Benefits:

- Retain the robustness benefits of redundancy
- Minimize the bandwidth cost
- Application can signal the importance of data. (Fine-grained control)

# A Case Study of RPT

## Redundant Packet Transmission (RPT)

- Send multiple copy of the same packet.
- Send every packet  $r$  times.
- Applied to live video in RE networks.

### Redundant Packets Transmission

Hop-by-hop RE networks

Partially content-aware networks

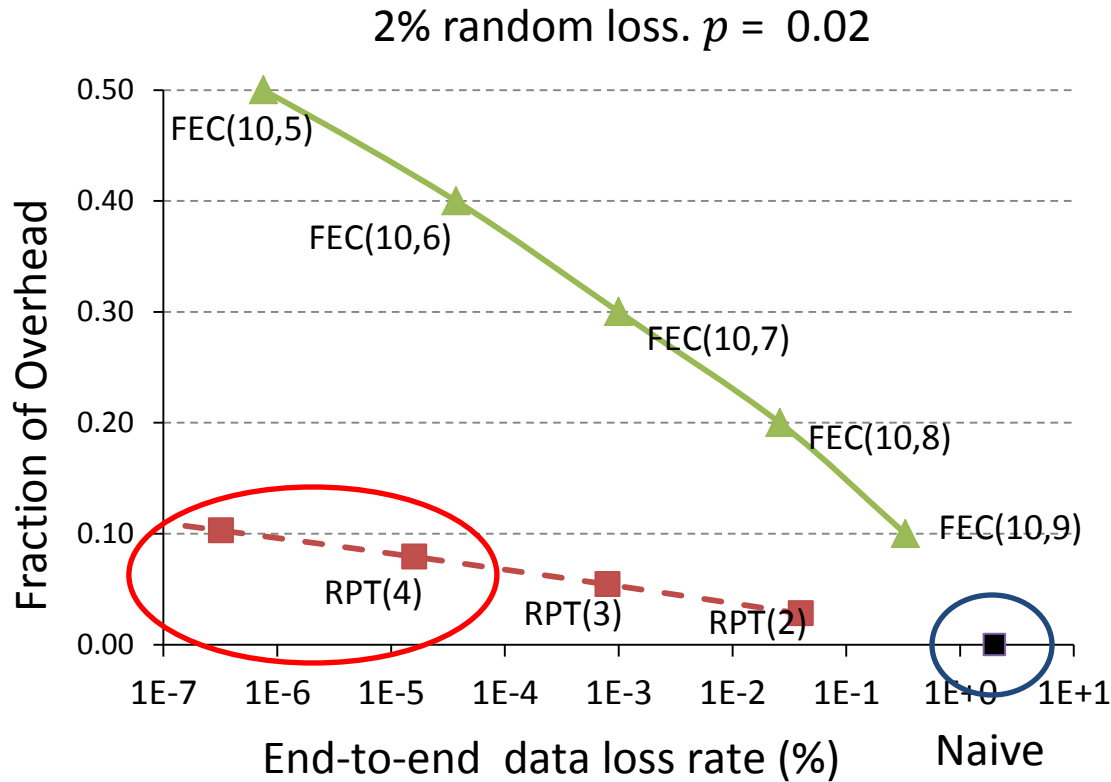
SmartRE networks

Networks with link-layer loss

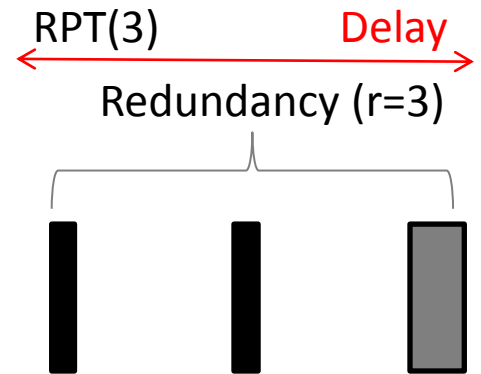
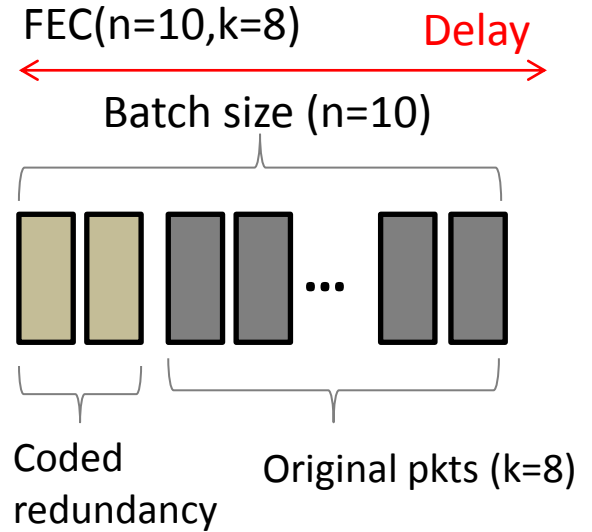
Content Centric Networks (CCN)

Time-critical  
communication

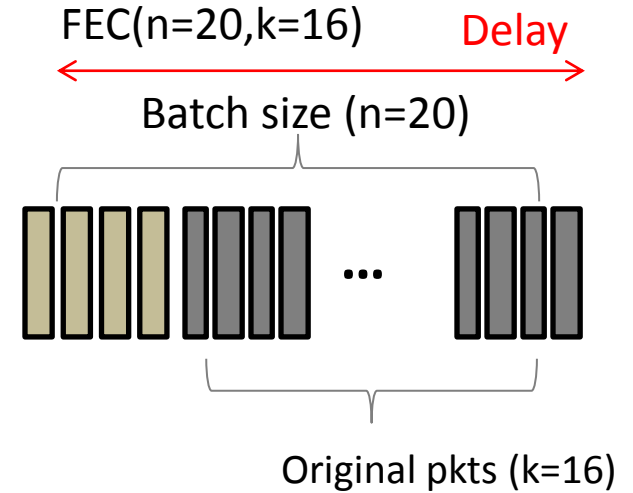
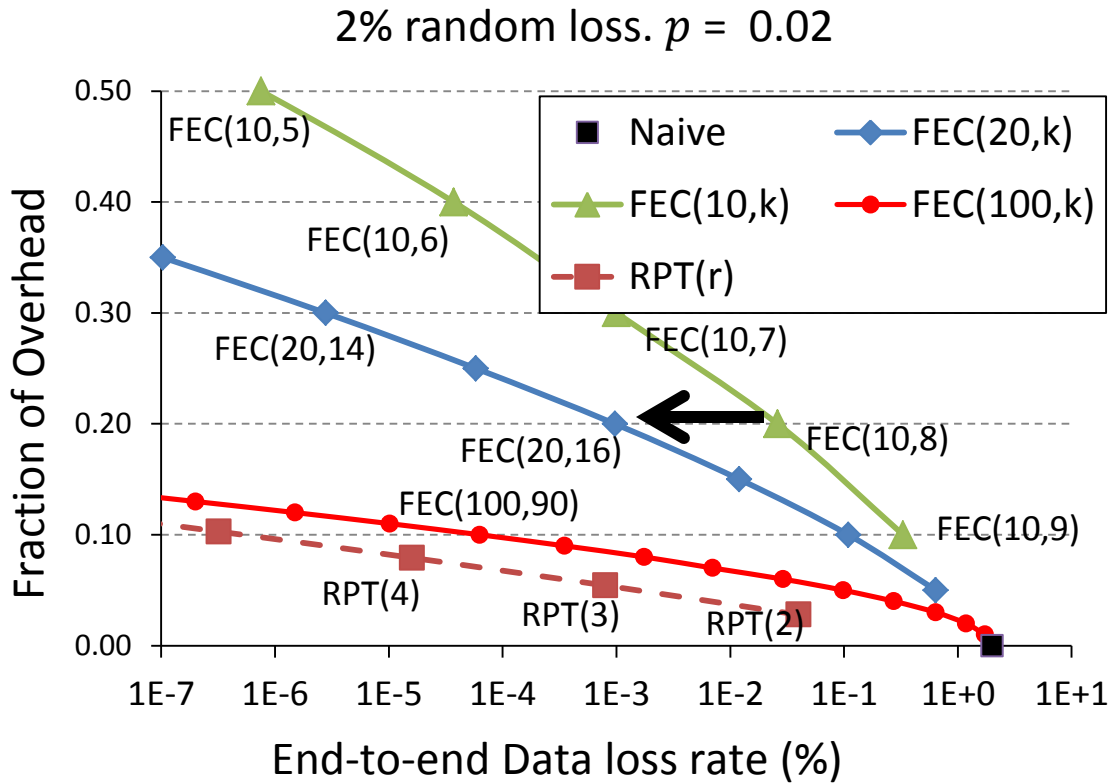
# Analytical Comparison with FEC



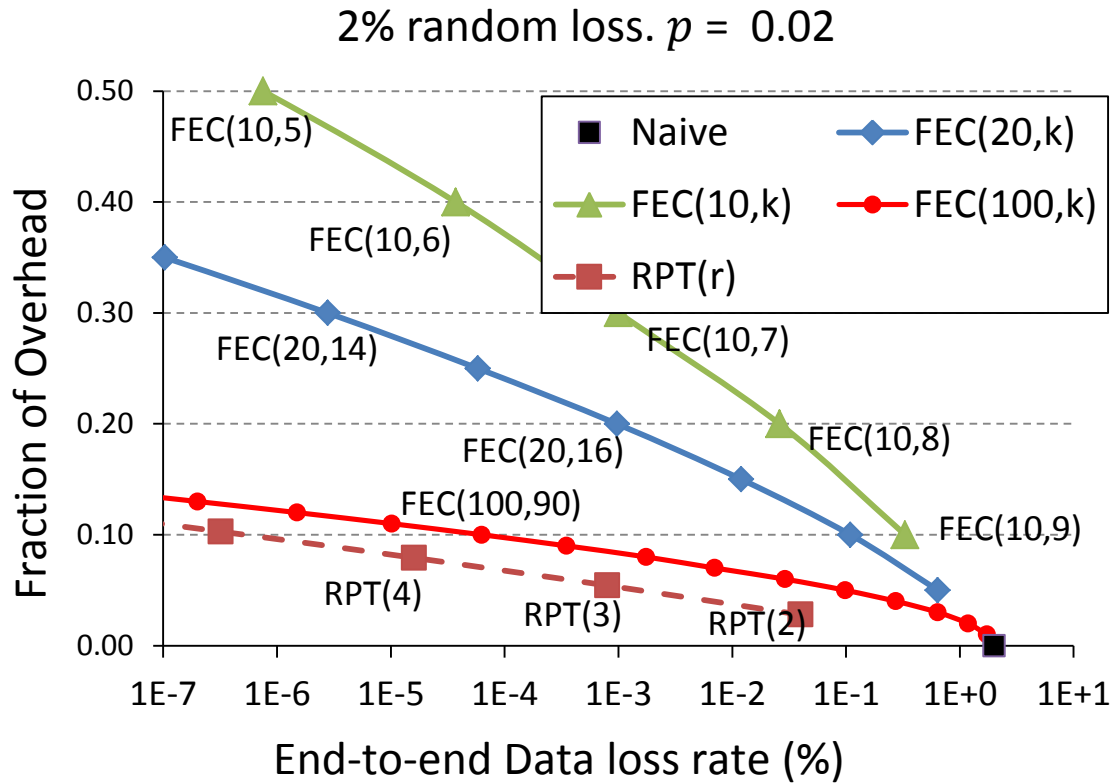
**Naive**  
**2% data loss**  
**0 overhead**



# Analytical Comparison with FEC



# Analytical Comparison with FEC



Scheme	Max Delay@ 1Mbps
FEC(10,7)	168 ms
FEC(20,16)	300 ms
FEC(100,92)	1300 ms
RPT(r)	<b>Tunable</b>

Skype video call 128~300kbps  
 Skype (HD) 1.2~1.5Mbps

# Experimental Evaluation

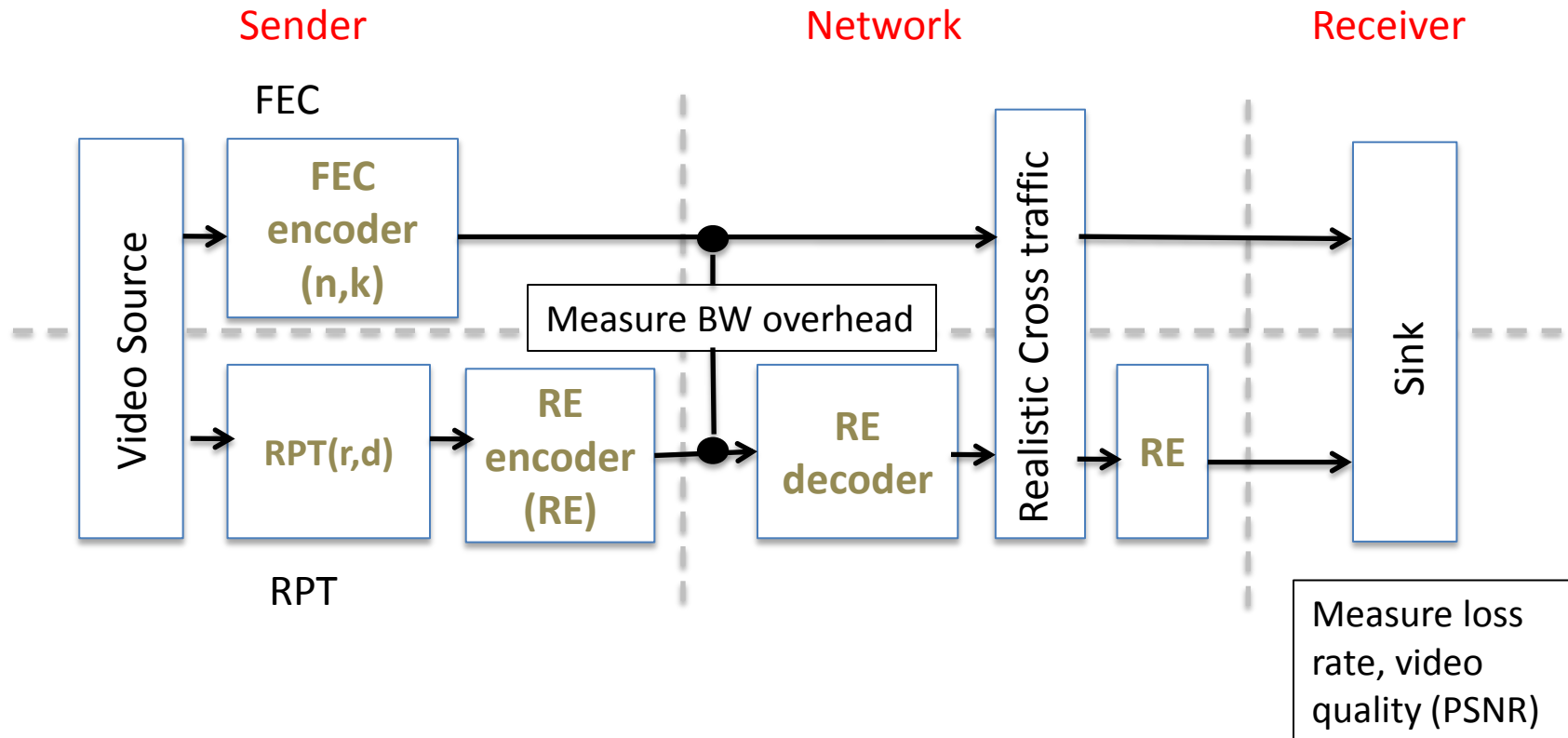
- Thorough evaluation on 3 different aspects of RPT
  - End user performance
  - Ease of use (parameter selection)
  - Impact on other traffic
- Methodology
  - Real experiment
  - Trace based experiment
  - Simulation



CIF: 352x288

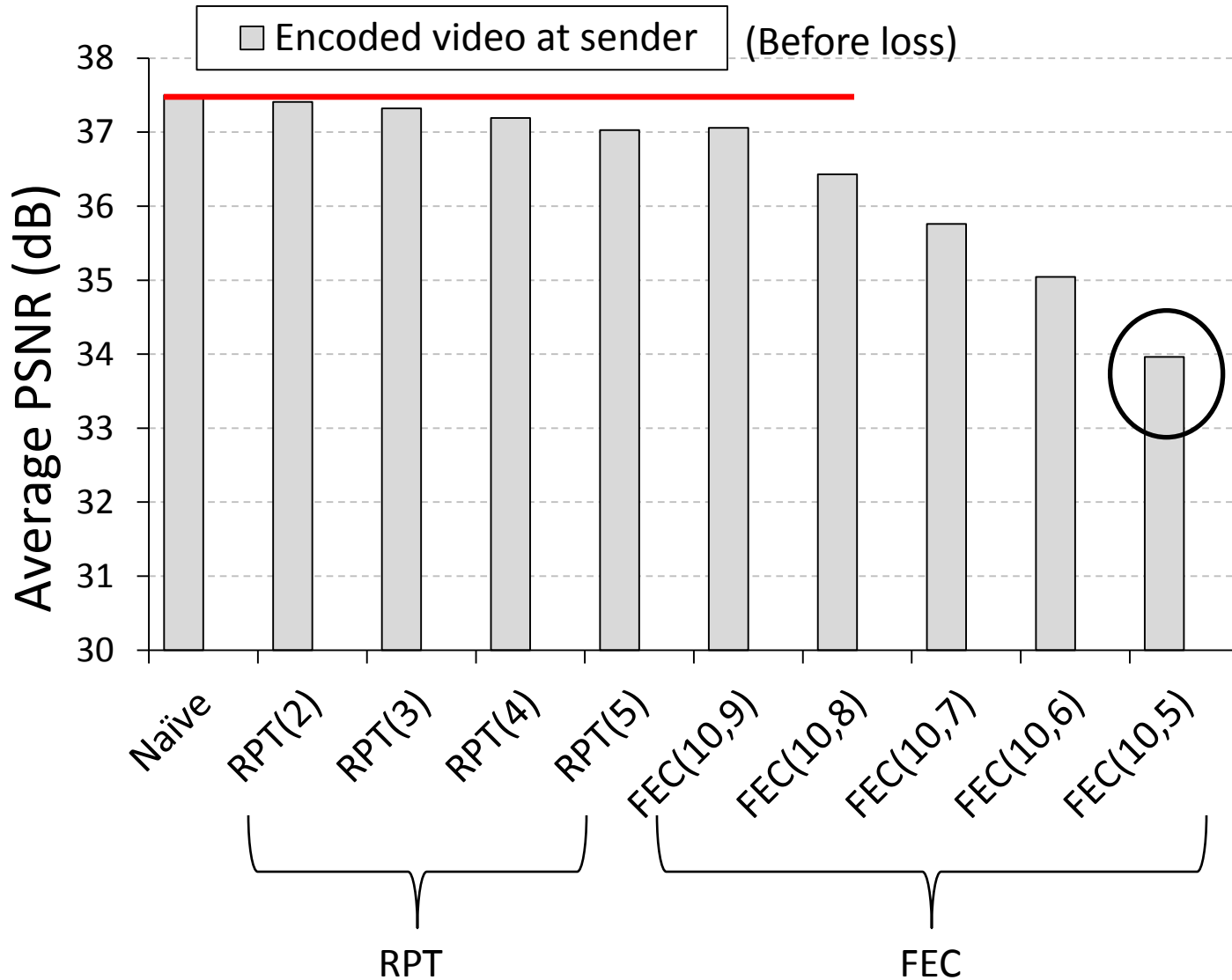
# Evaluation Framework

- RE router implementation (Click, NS2)
- Video quality evaluation using evalvid

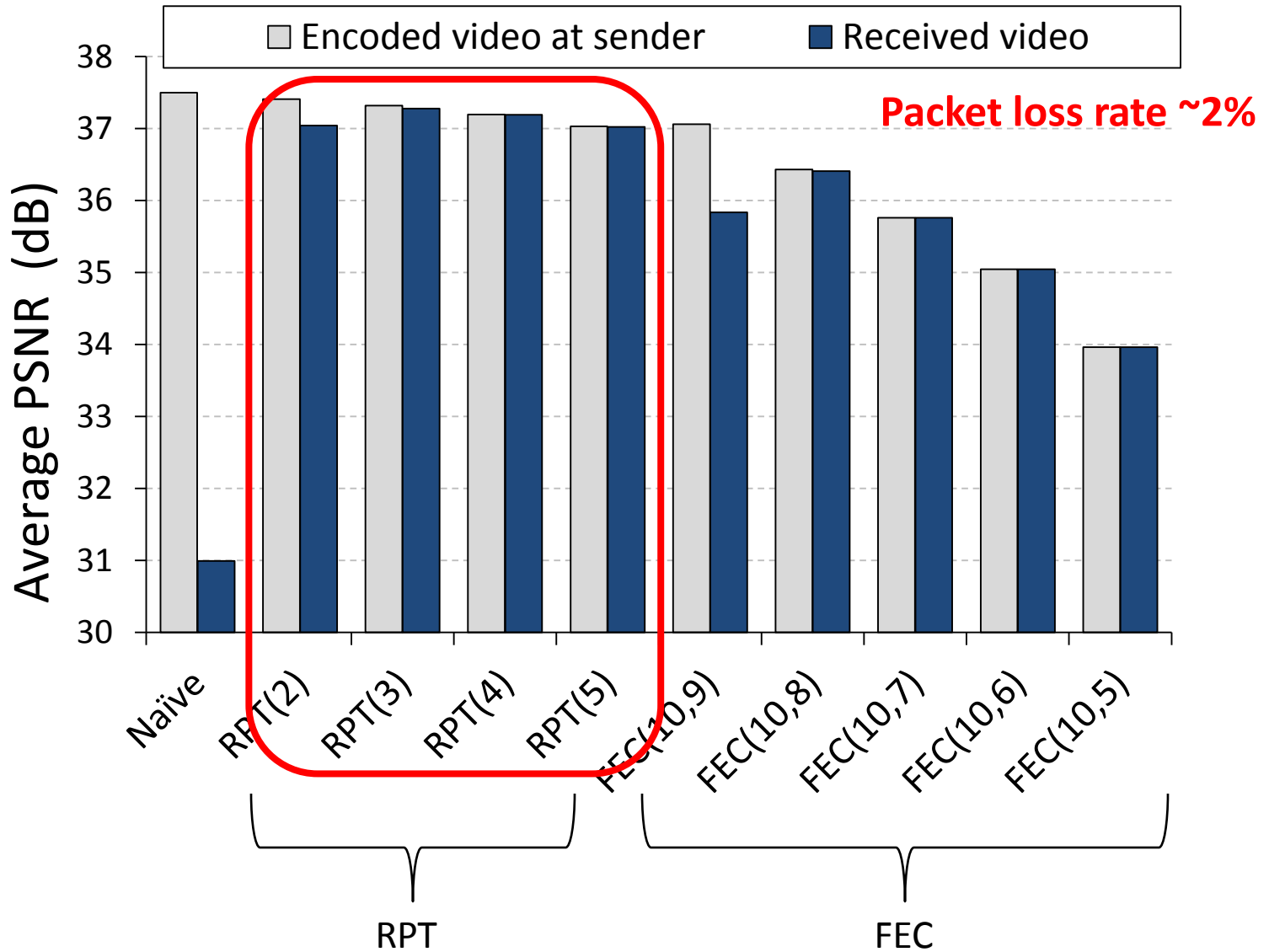




# E2E Performance: Video Quality



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RPT(3) Overhead ~6%

FEC(10,9) Overhead ~10%

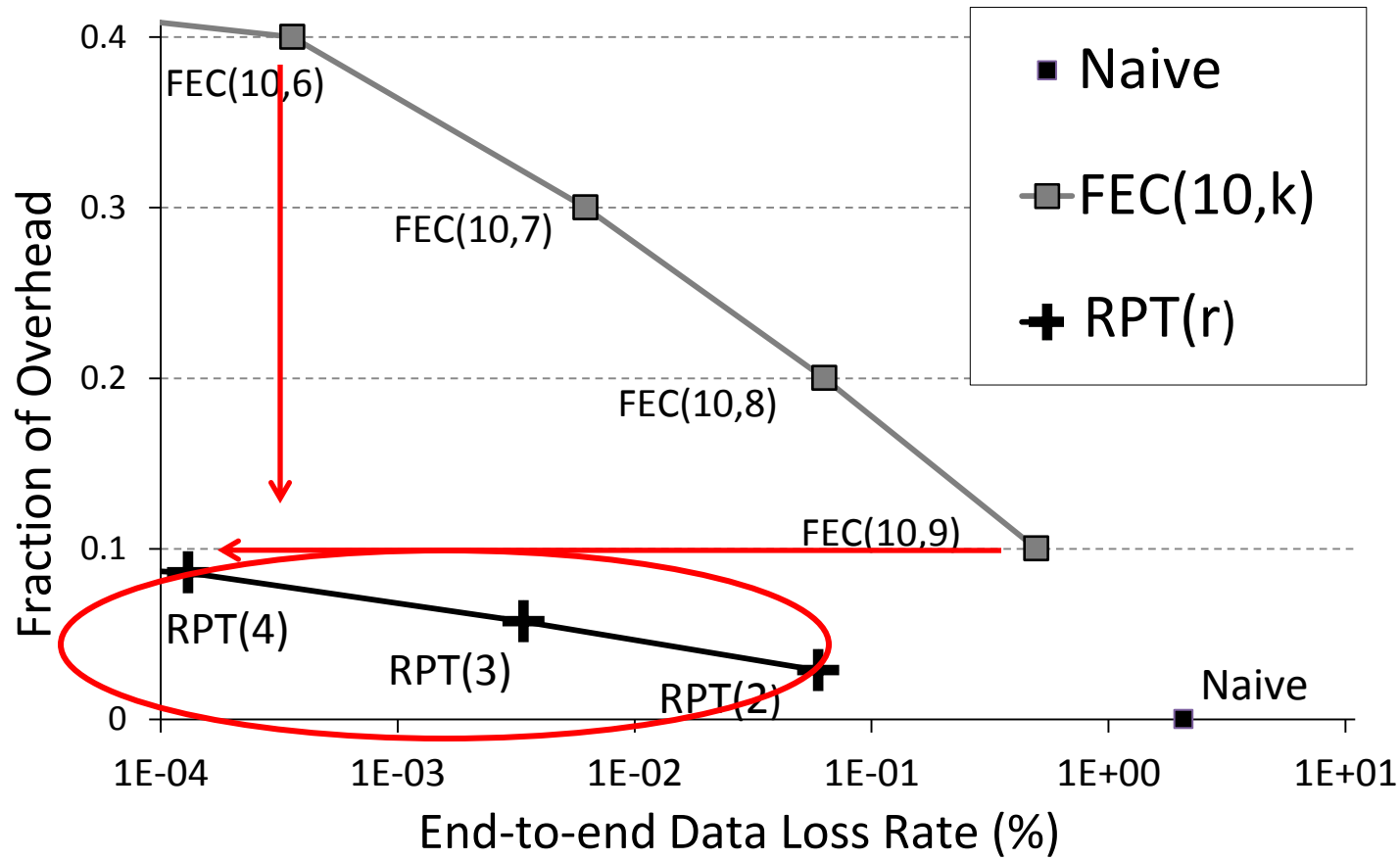
Naïve UDP



1.8dB ~ 3dB difference in quality

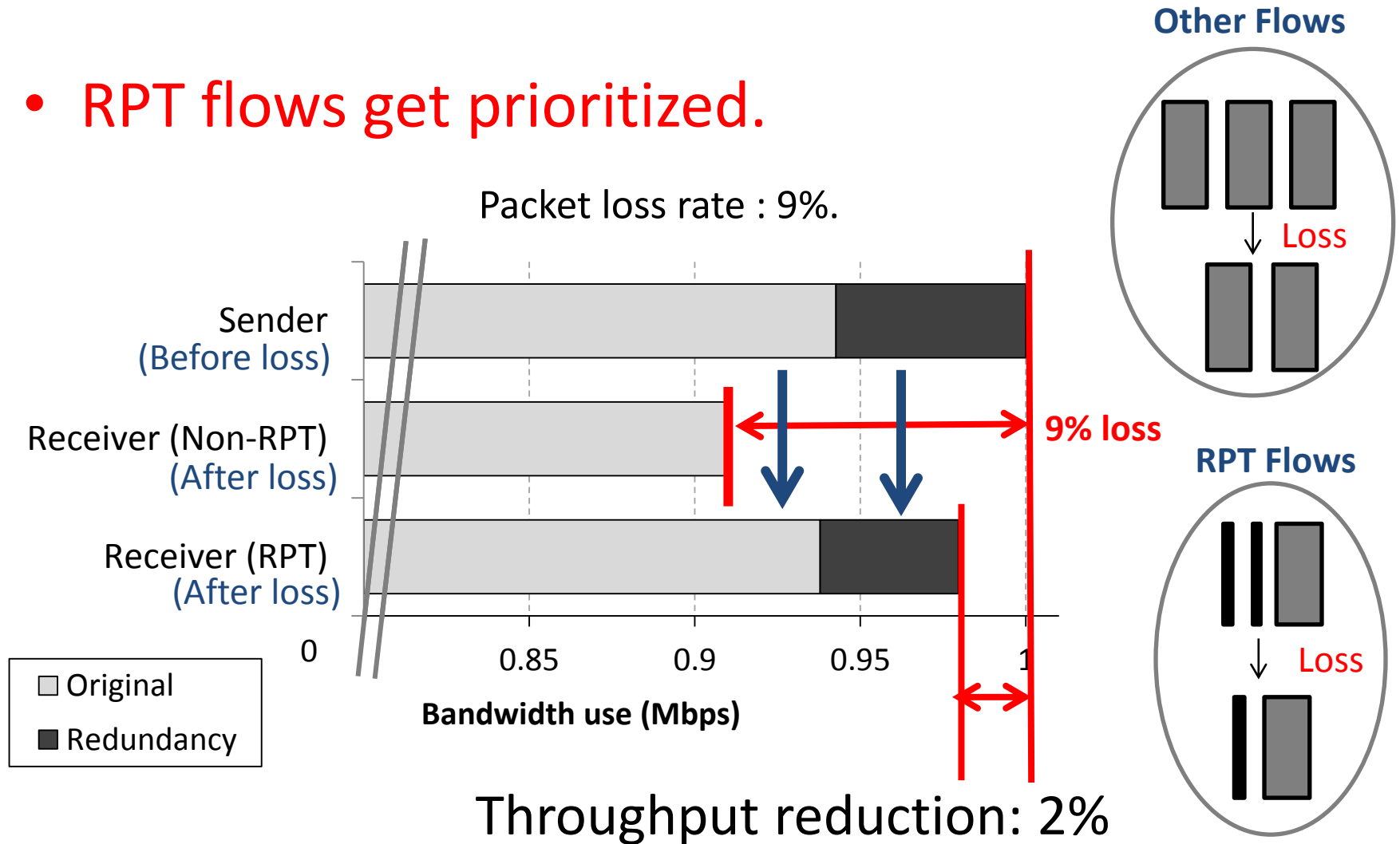
# E2E Performance: overhead and robustness

Packet loss rate ~2%



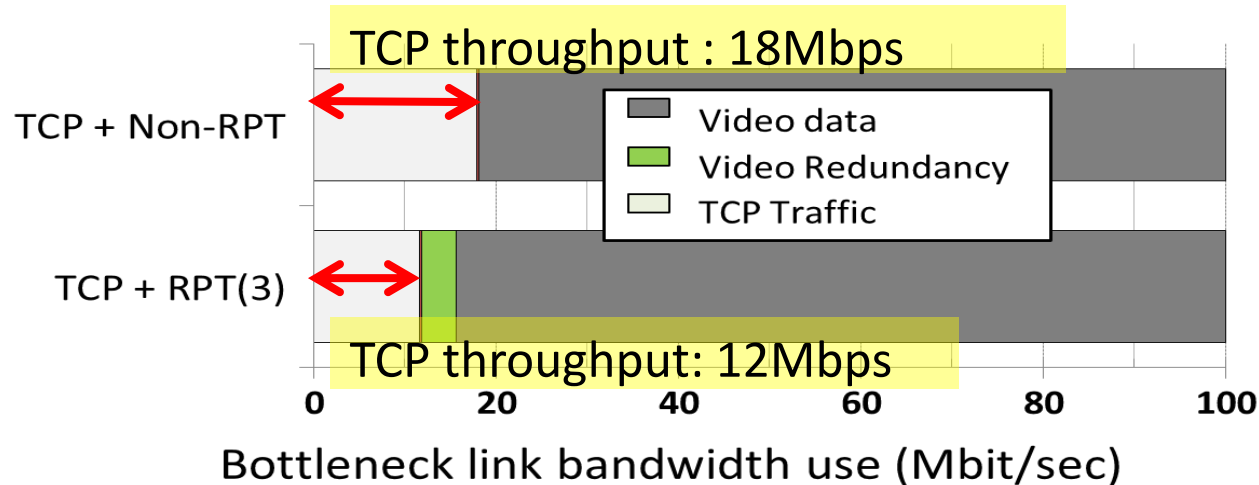
# Impact on other traffic

- RPT flows get prioritized.



# Is flow prioritization a problem?

- **Not a problem:** Important flows should be prioritized.
- **Problem: Unfair bandwidth allocation**



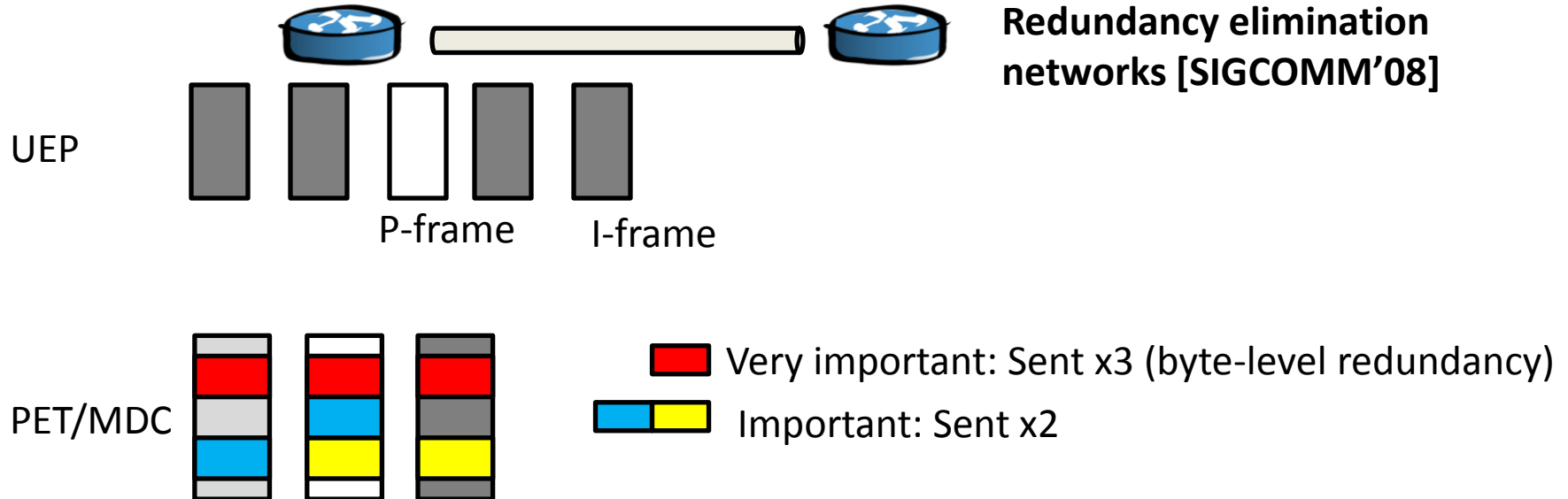
- **How do provide fairness and robustness at the same time?**
  - Core problem: RPT flows are not reacting to congestion.  
→ Apply TCP-friendly rate control to RPT.
  - Challenge: correctly accounting for possible changes in loss pattern

# Other results in the paper

- Demonstration of RPT in a real-world setting
  - E.g., Emulated corporate VPN scenario
- Trace-based experimental results
- Detailed parameter sensitivity study
- Network safety (impact on the network)
- Design and evaluation of TCP-friendly RPT
- Strategies on other content-aware networks

# Generalized RPT

- Many sophisticated schemes are enabled by FEC.
  - Priority encoding transmission (PET), unequal error protection (UEP), multiple description coding (MDC)
- ➔ Prioritization within a flow for graceful degradation of quality





# Conclusion

- Key Idea of RPT: Don't hide, expose redundancy!
- Key Features
  - High robustness, low overhead → user performance
  - Ease of use: parameter selection, per-packet redundancy/delay control
  - Flow prioritization
- Applicability
  - Applies to delay-sensitive communications in content-aware networks in general.

