

Adaptive FEC for Congestion Control

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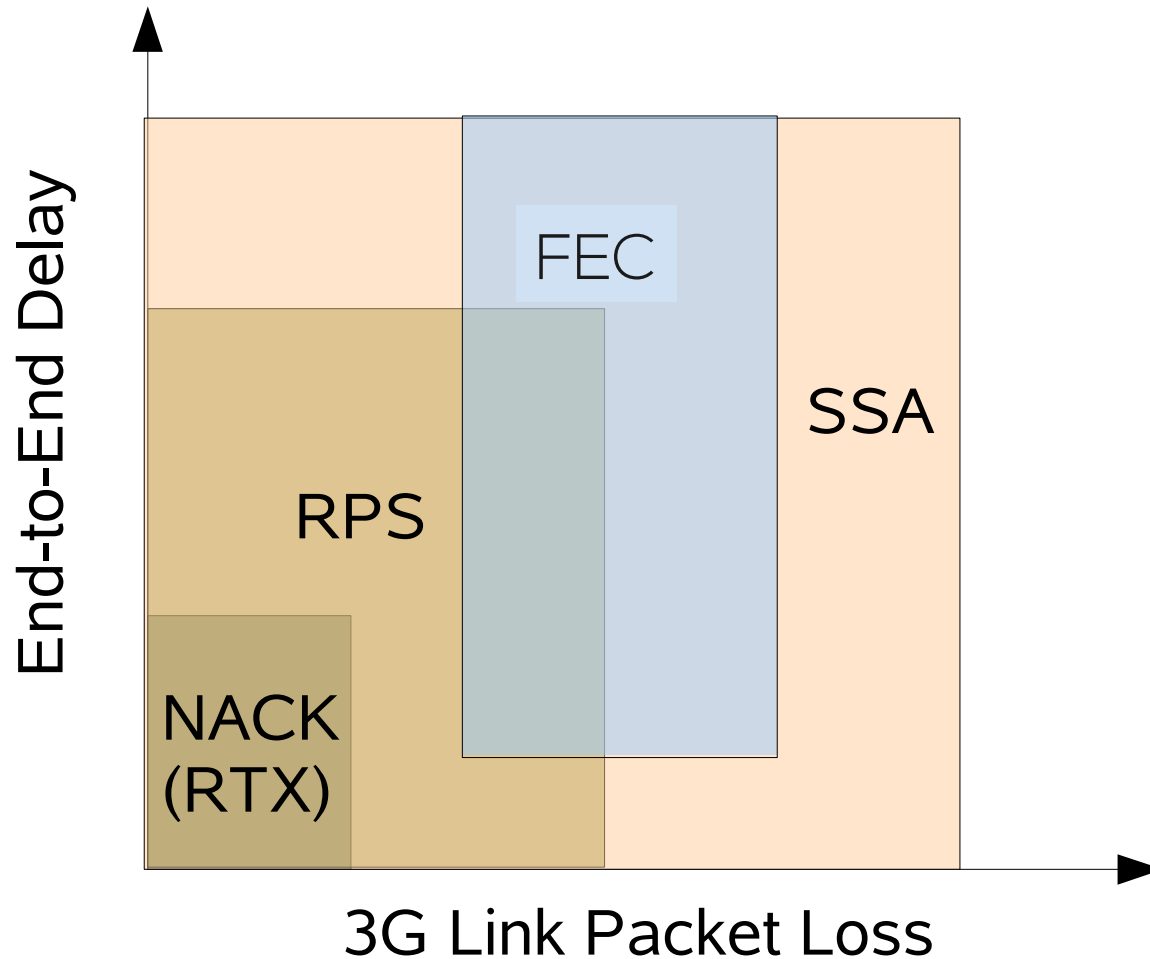
IETF 91, Honolulu,
09. November 2014

<https://tools.ietf.org/html/draft-singh-rmcat-adaptive-fec-01>

IPR

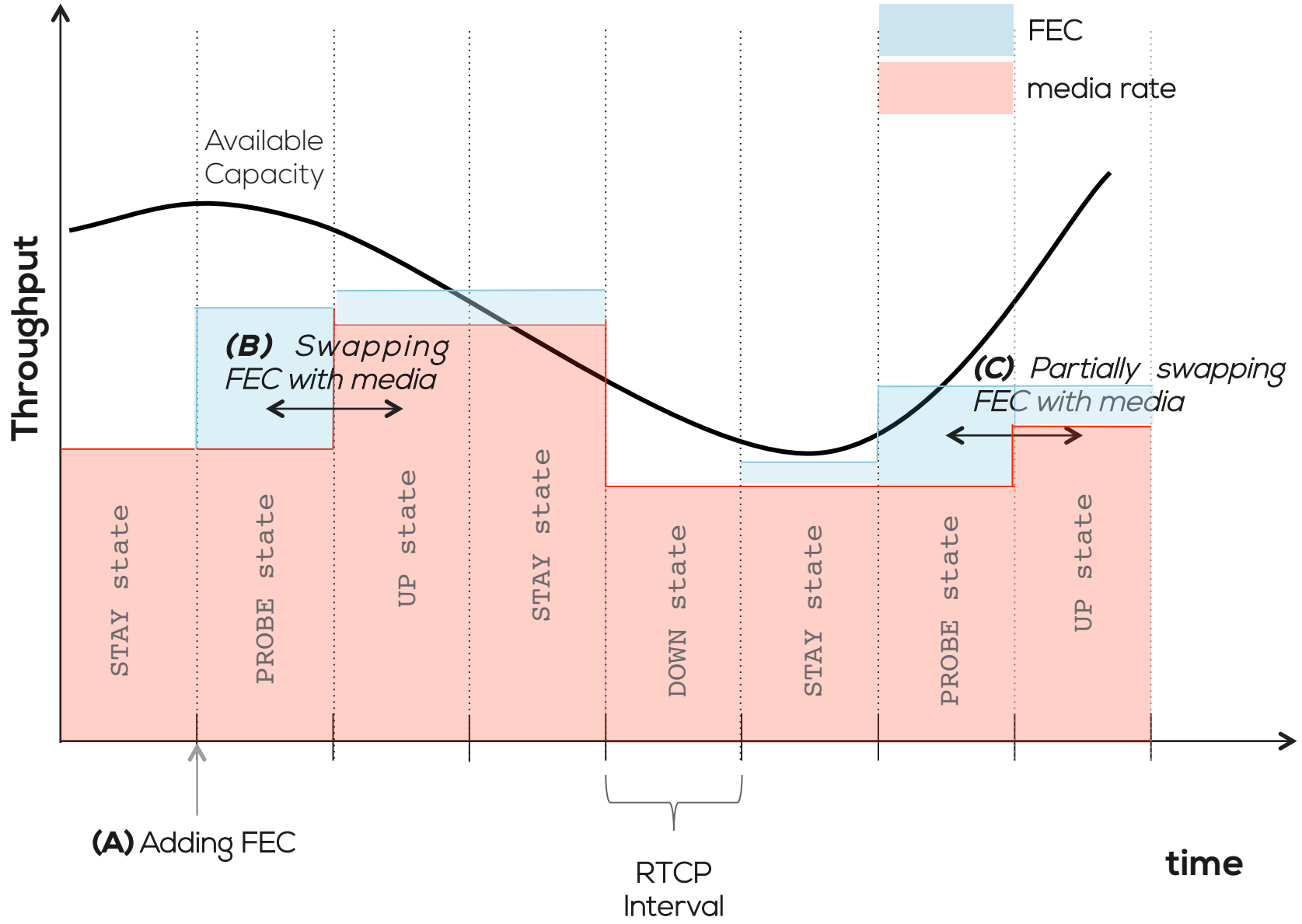
- 2 IPR disclosures
 - Nokia
 - Polycom

Error Resilience

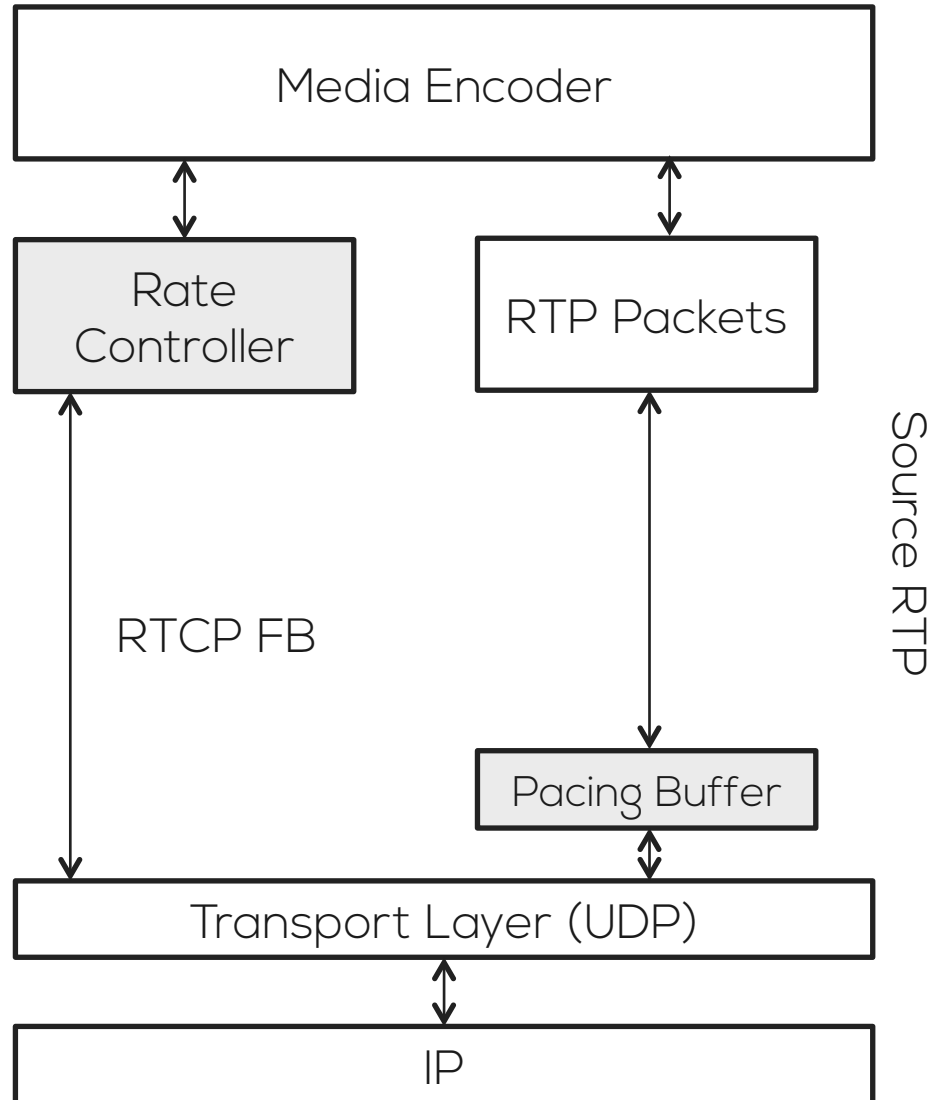


SSA: Adaptive packet sizes
RPS: reference picture selection

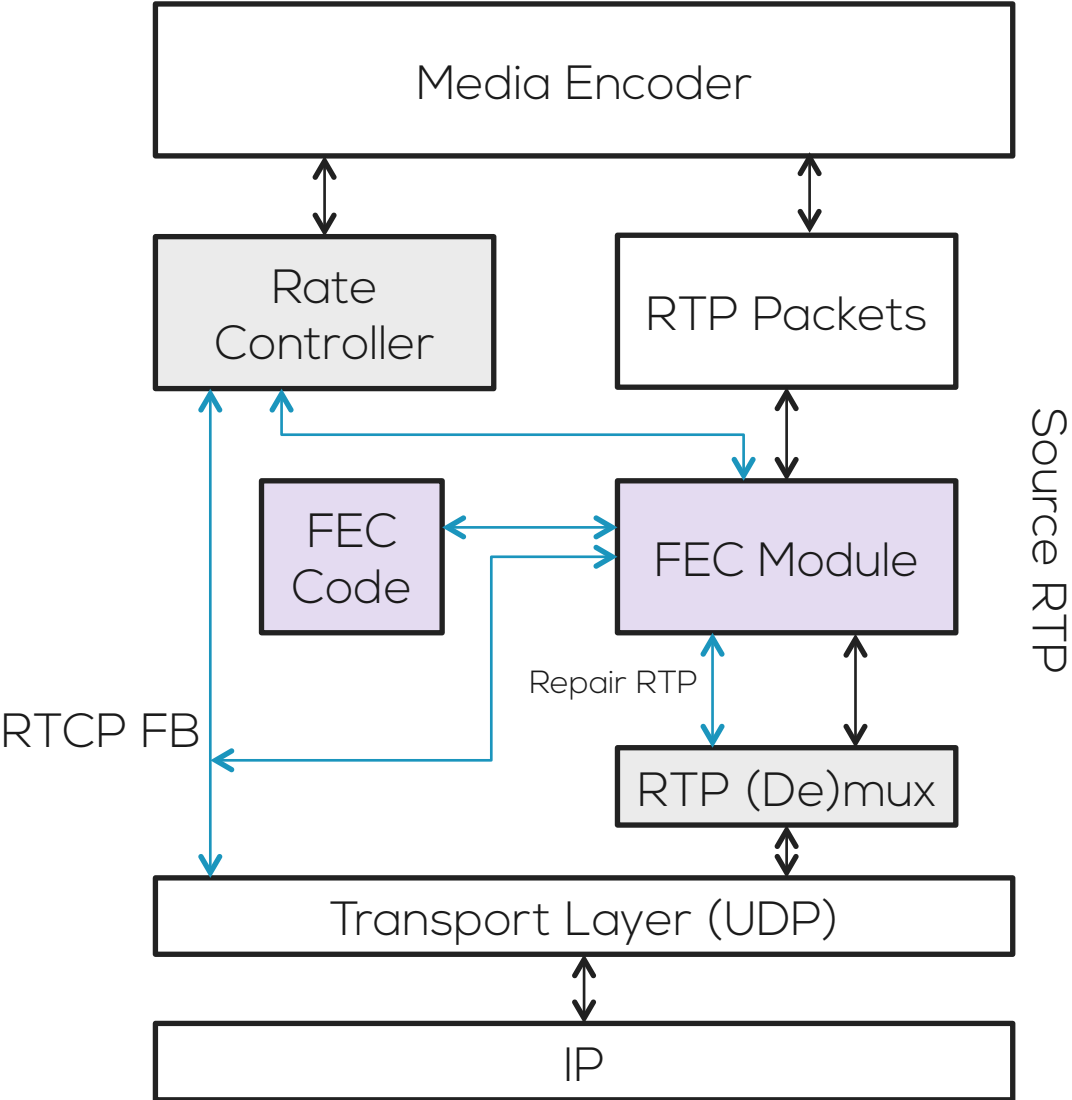
CONCEPT



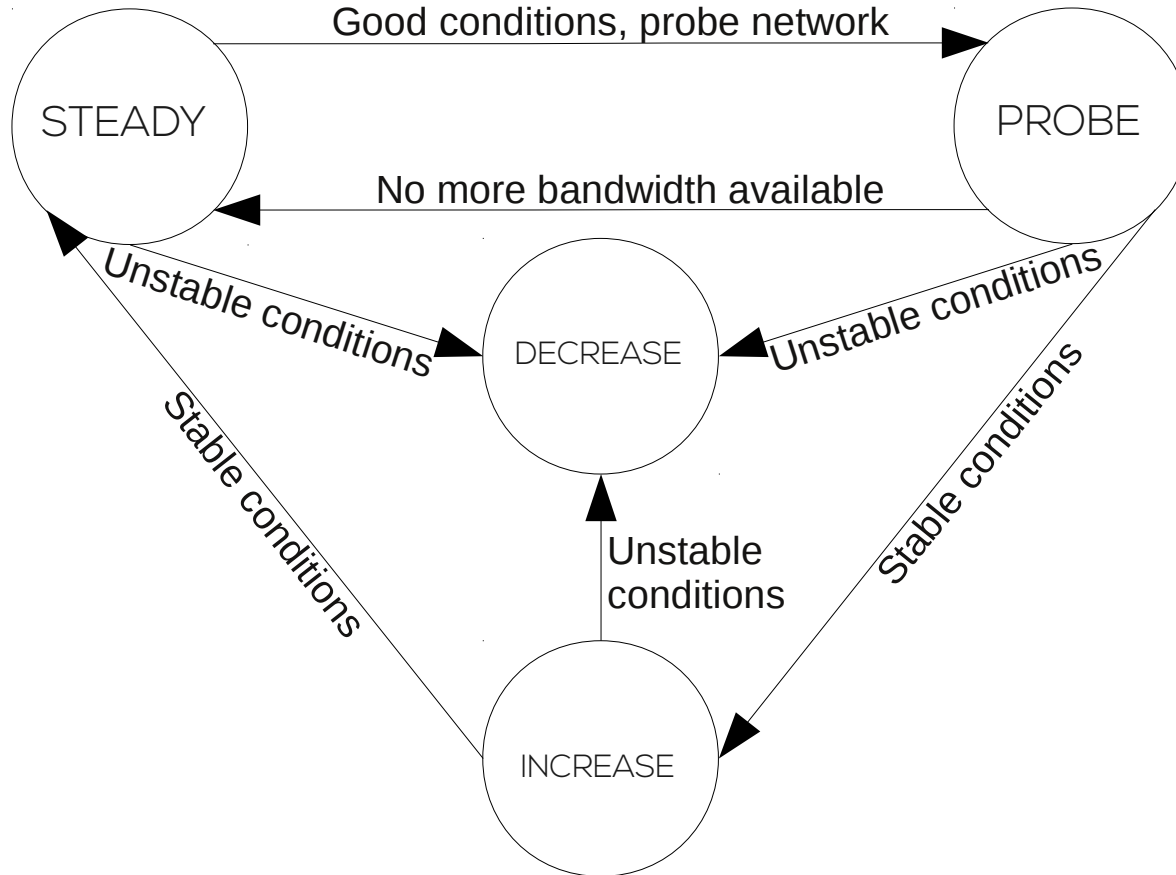
CC Framework



RFC 6363: FEC Framework



State Machine



FEC Scheme

- Currently, using
 - parity, 1- or 2-d interleaved XOR (for burst loss)
 - There is discussion in RTCWEB on FEC schemes.
- But,
 - It could be applied to other schemes

RTCP Feedback

- RLE of Post-repair (RFC5725)
- RLE of loss packets (RFC3611)
- RLE of discarded packets (RFC7097)

- Packet count of lost and repaired packets
 - (draft-ietf-xrblock-post-loss-repair)

Calculating goodput

- Receiver reports goodput:
 - Goodbytes in reporting interval =
bytes received – bytes discarded

OR

- Sender calculates goodput:
 - Goodbytes in reporting interval =
bytes sent – bytes lost – bytes discarded


RLE reports

Undershoot

- Congestion reported
 - Sender calculates duration of congestion =
HSN when RTCP scheduled – HSN when congestion detected
 - undershoot bytes = sending rate * duration/8
 - New rate = Goodbytes – undershoot bytes [1]

OR

- Delta = Sending rate – goodput
- New rate = sending rate – 2 x delta [2]

[1] <http://www.netlab.tkk.fi/~jo/papers/2012-videw-rate-control.pdf>

[2] <http://www.netlab.tkk.fi/~varun/nagy2014mmsys.pdf>

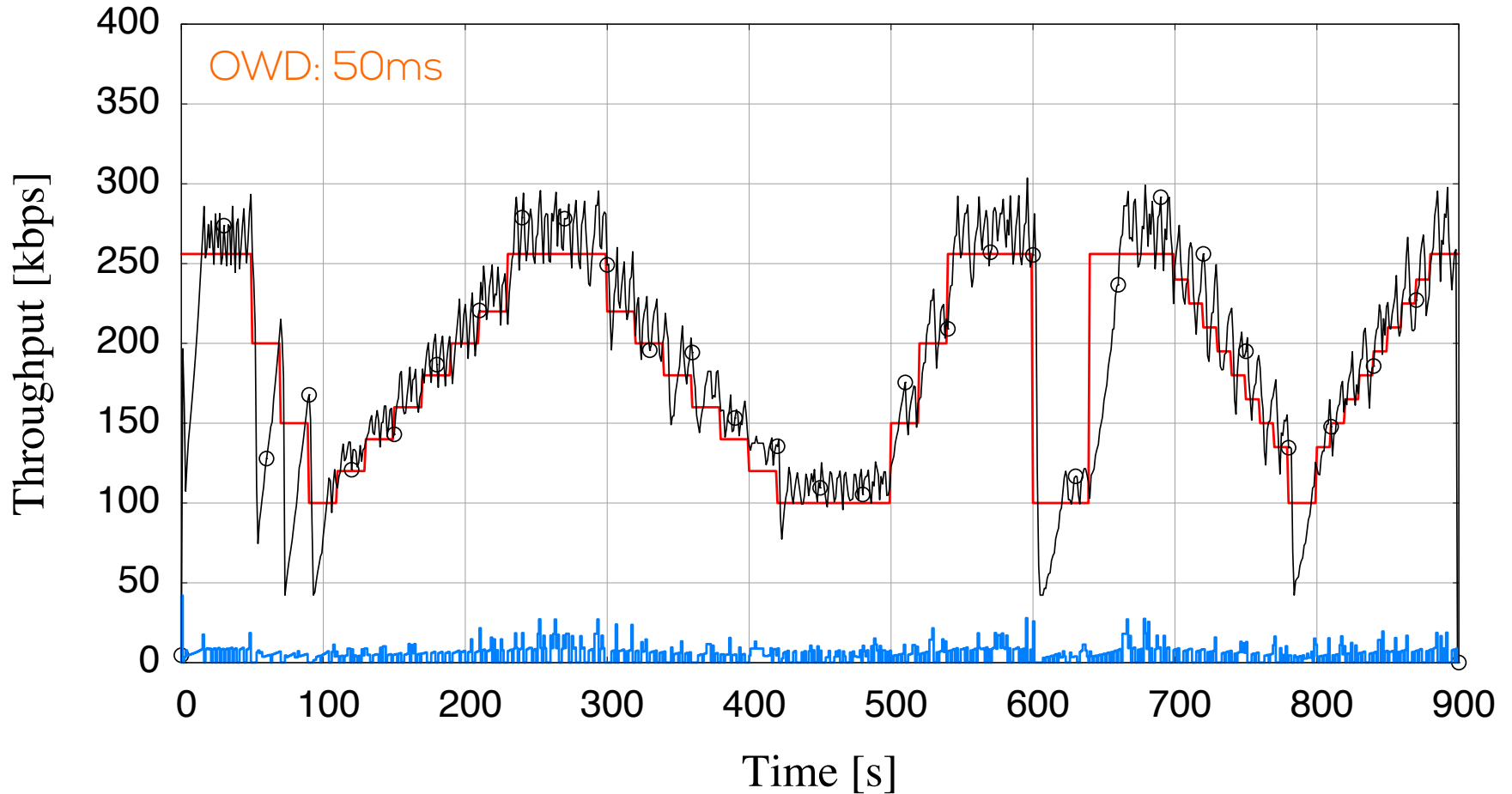
Typical results

- ~90% recovery in time for decoding when FEC interval was short
 - Fewer packets protected
- ~80% recovery in time for decoding when FEC interval was long
 - More packets protected

Applicability

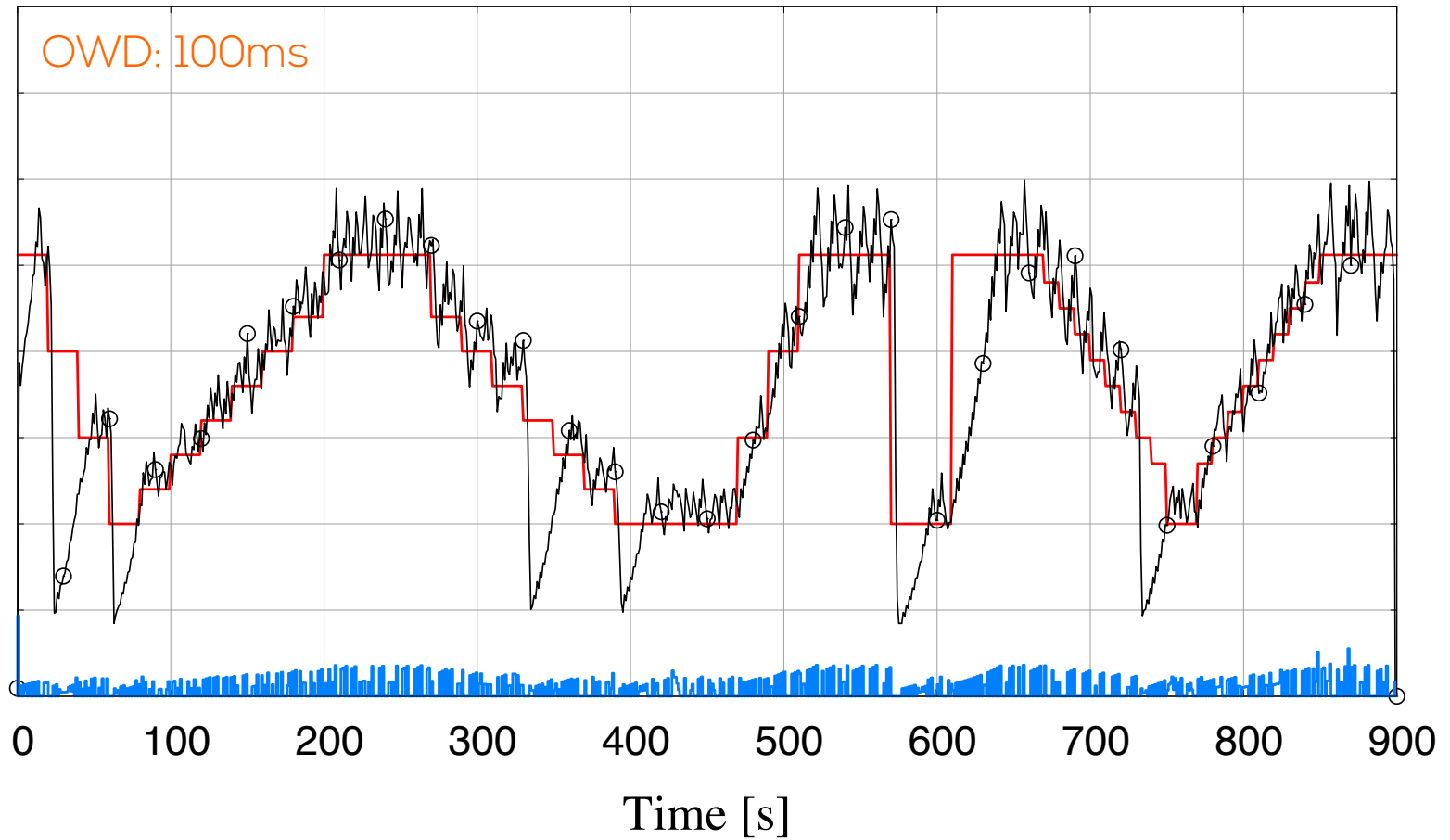
- Implemented over a delay-based congestion control
 - See paper for details
- However would like to generalize it.
 - Apply to SCReAM, GCC, ...

Evaluation (1/3)

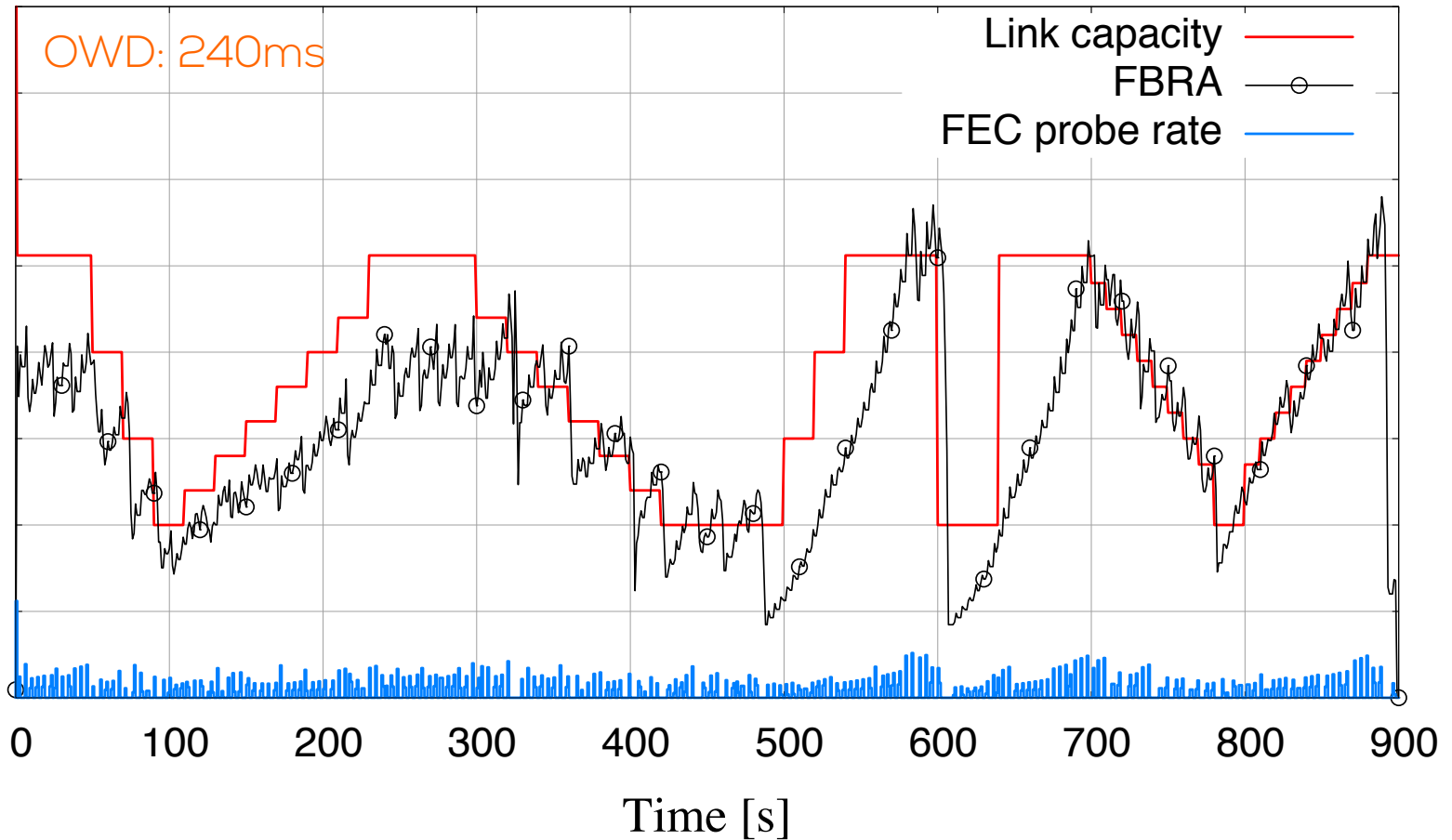


- Ns-2 simulation, Variable link capacity, Single flow on the link

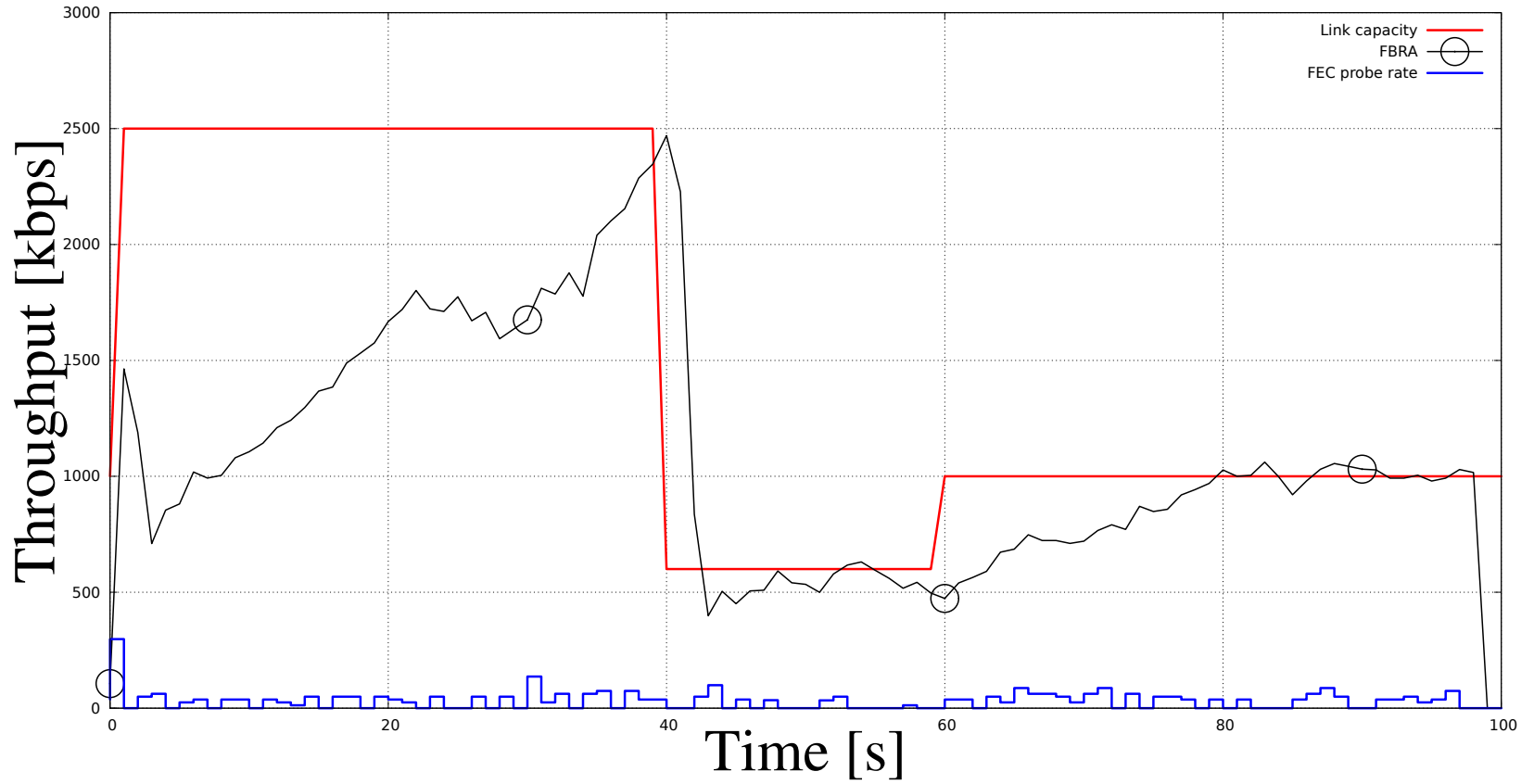
Evaluation (2/3)



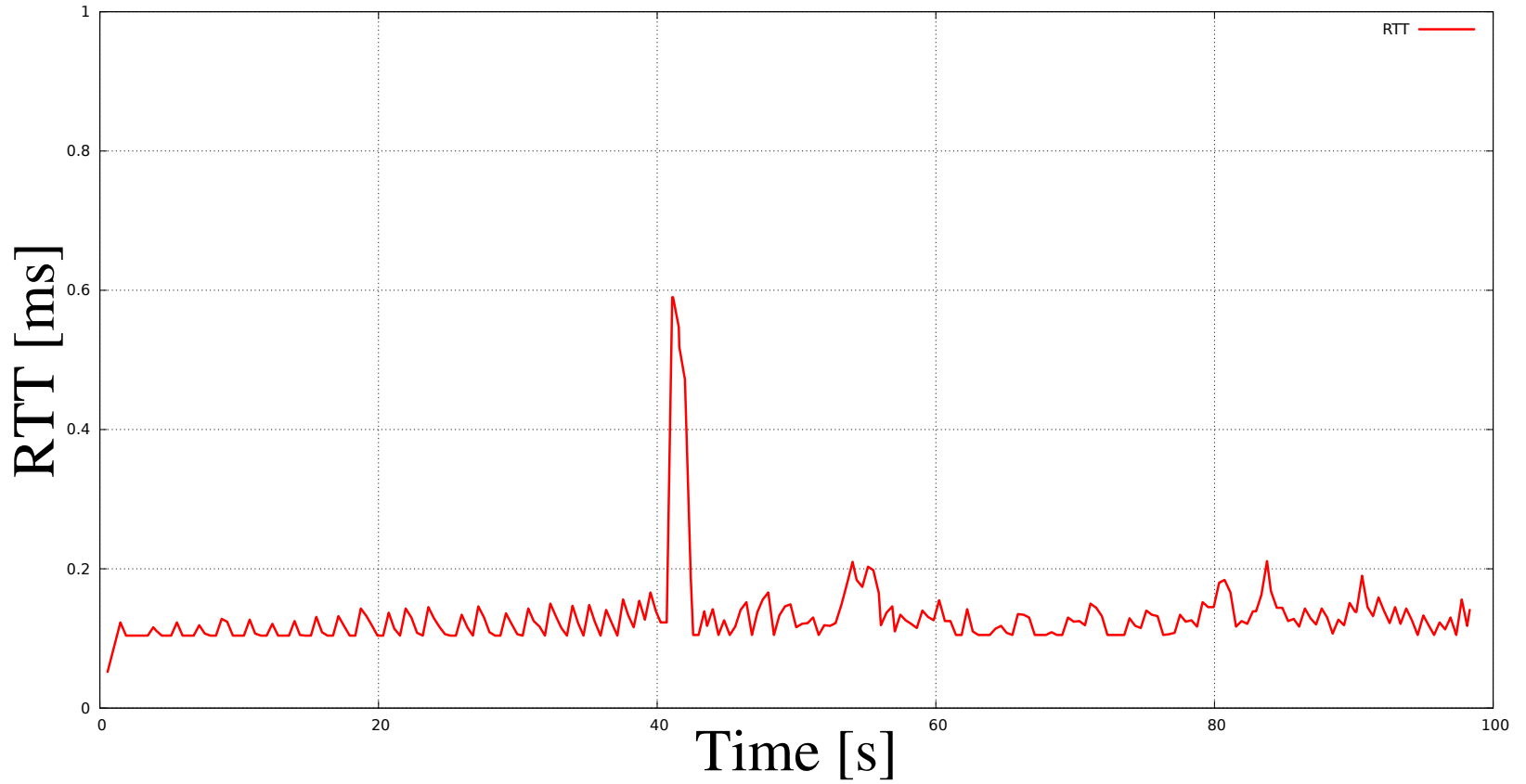
Evaluation (3/3)



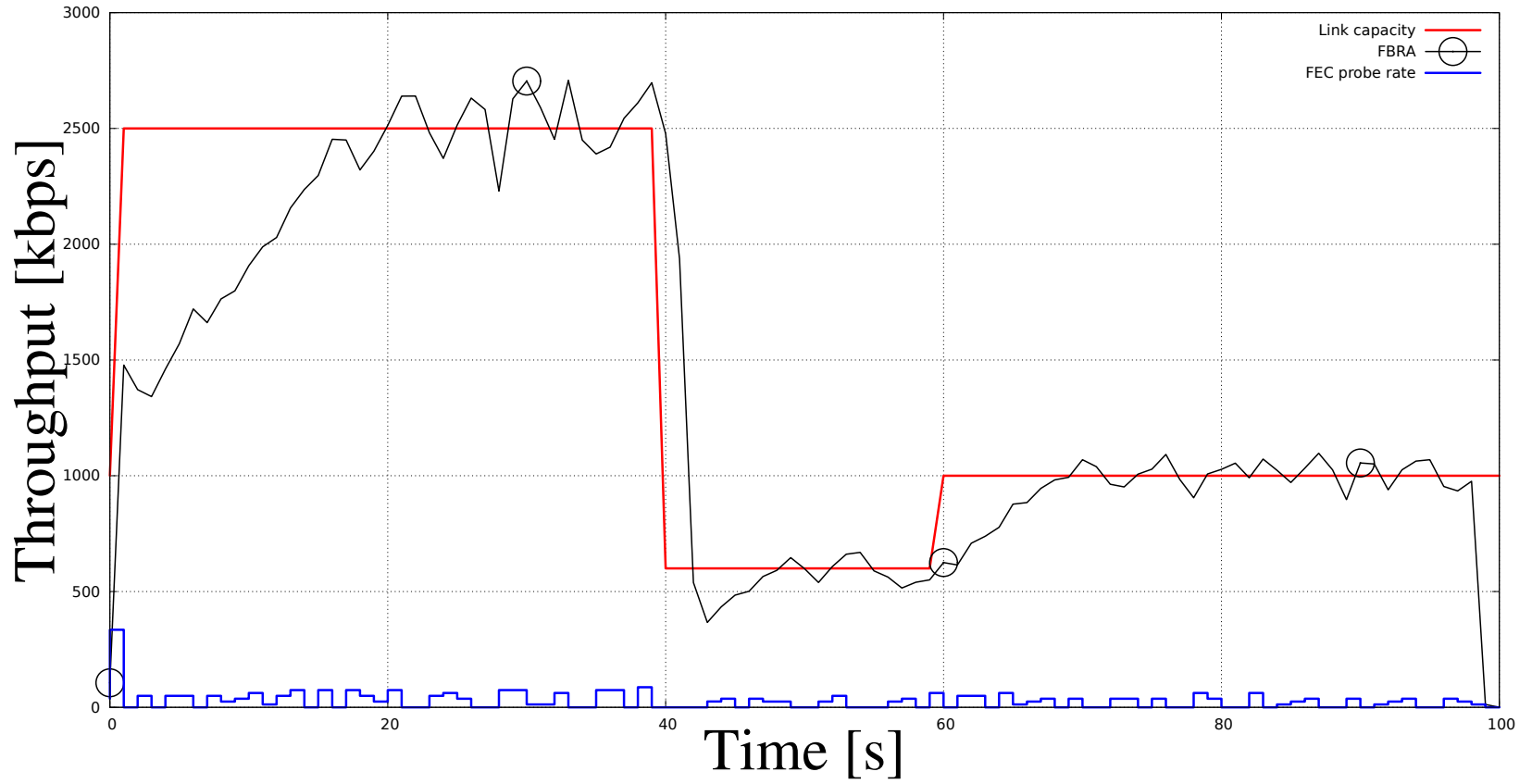
50ms



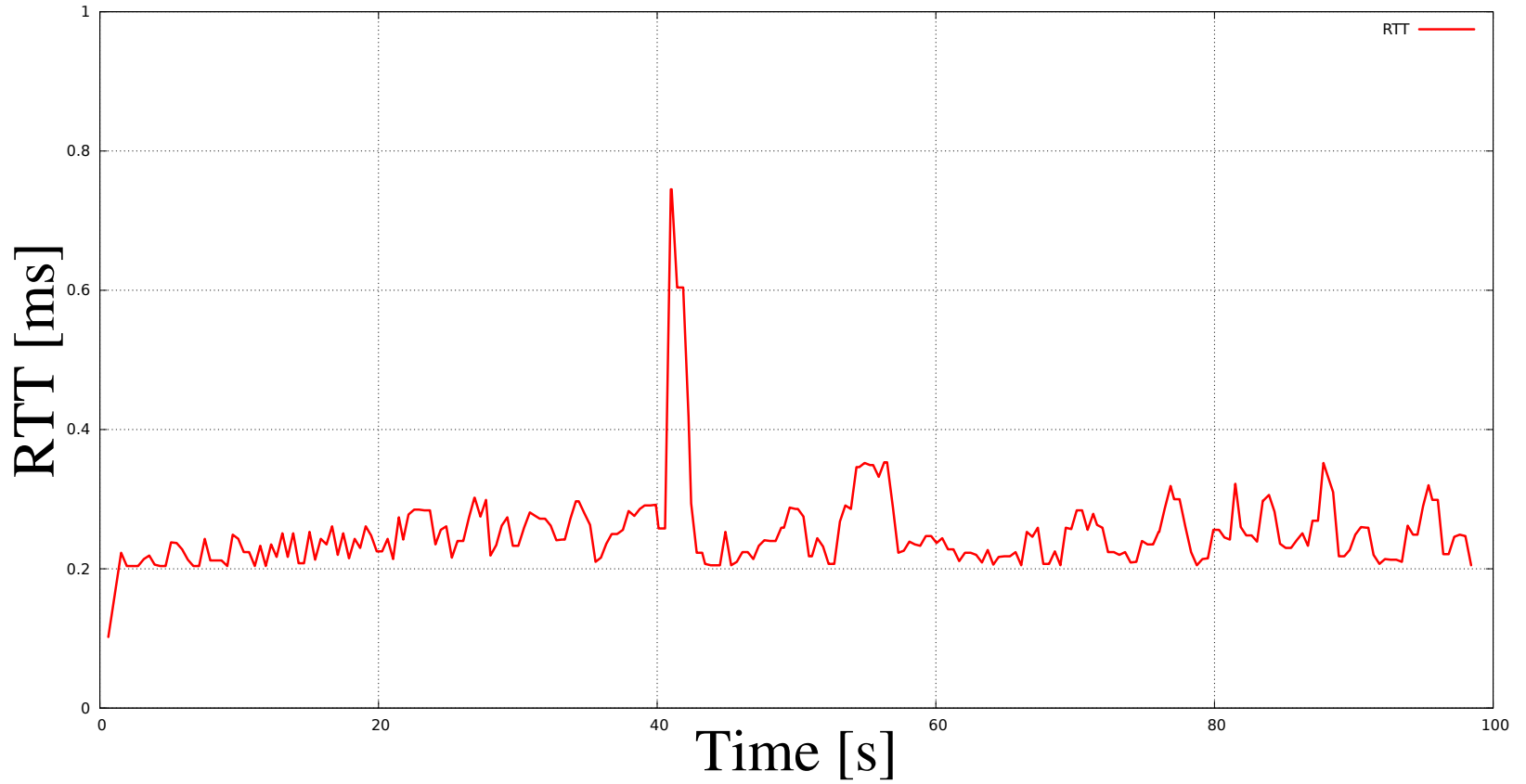
50ms



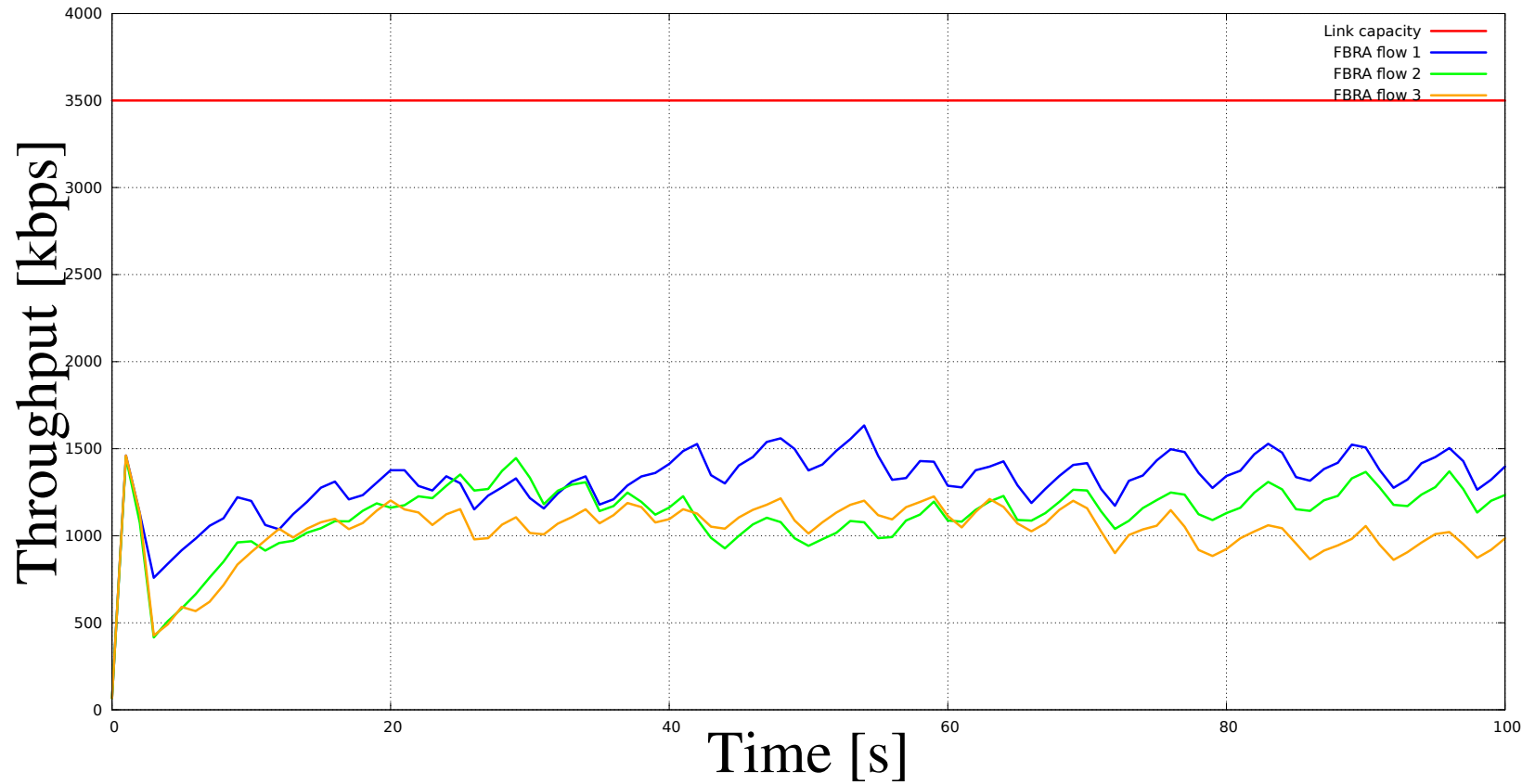
100ms



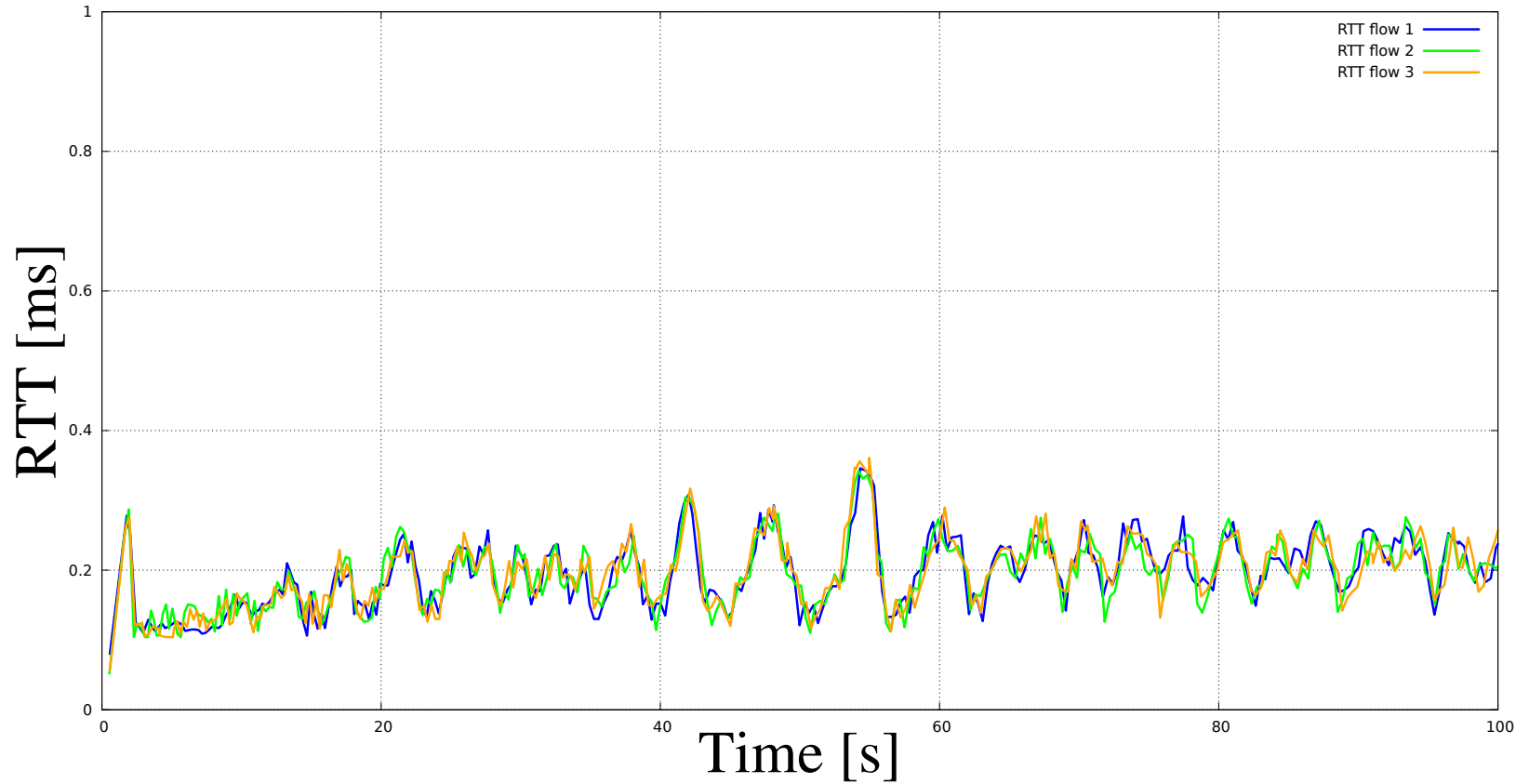
100ms



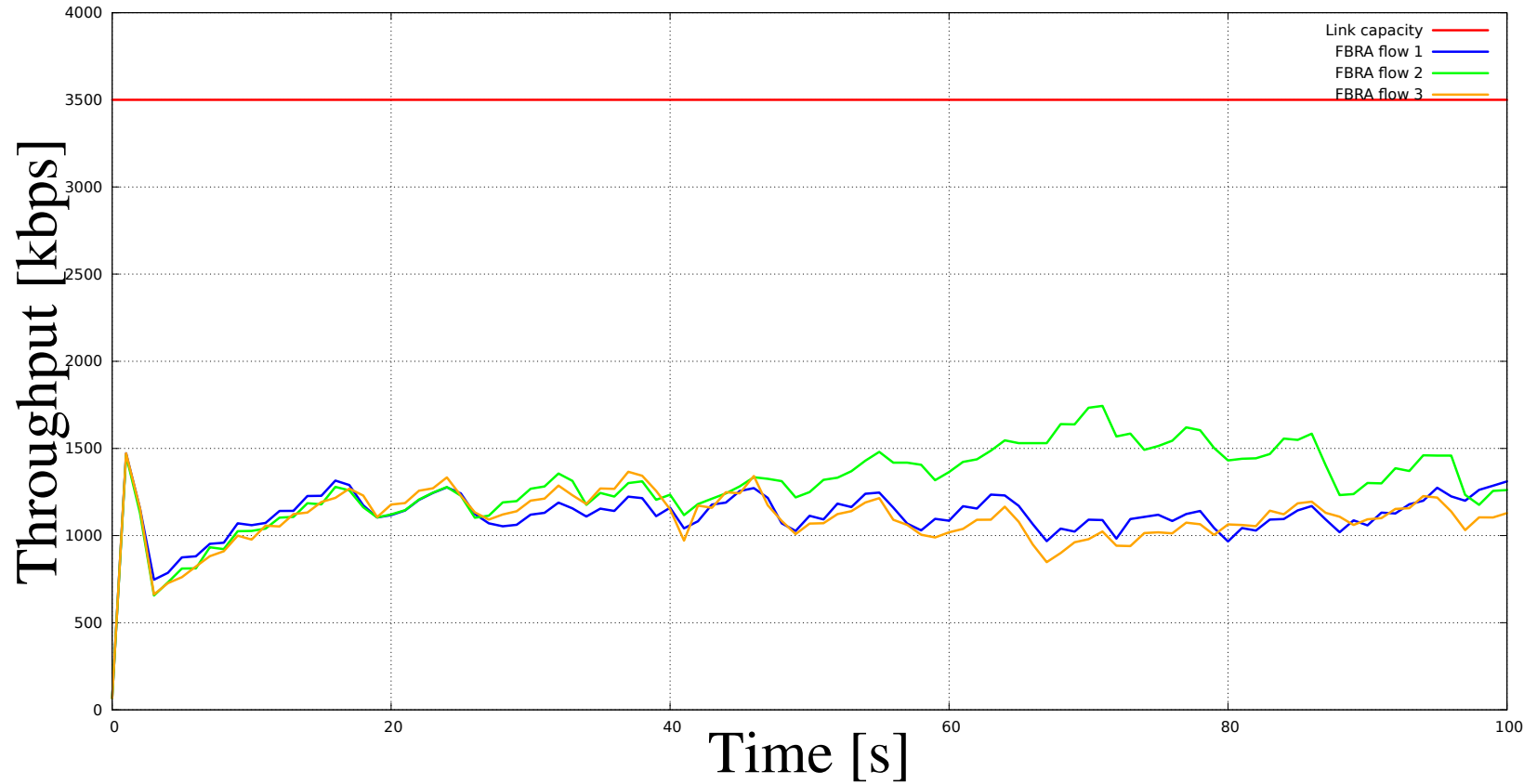
RTT fairness (50ms)



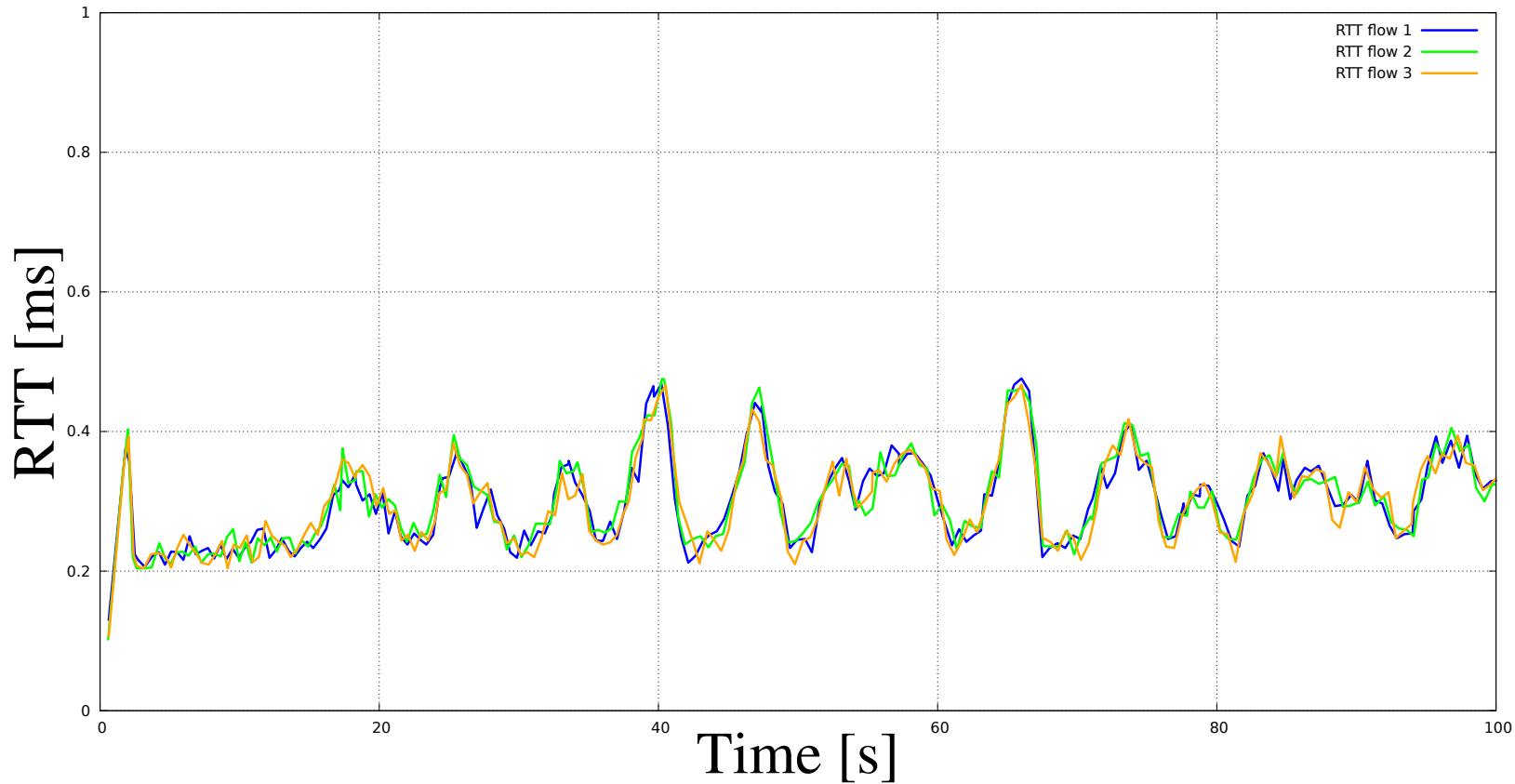
5.4: RTT fairness



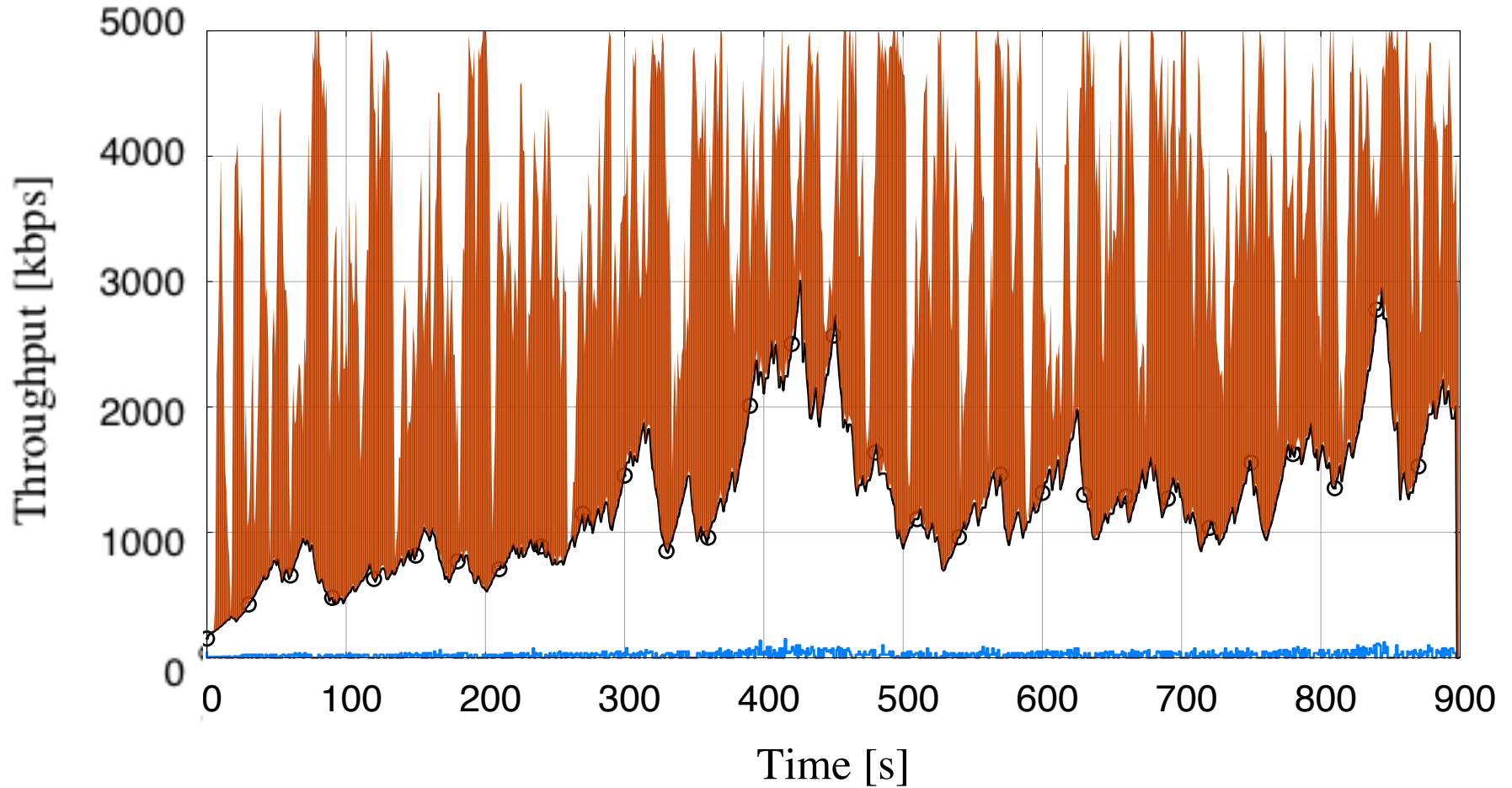
RTT fairness (100ms)



RTT fairness (100ms)

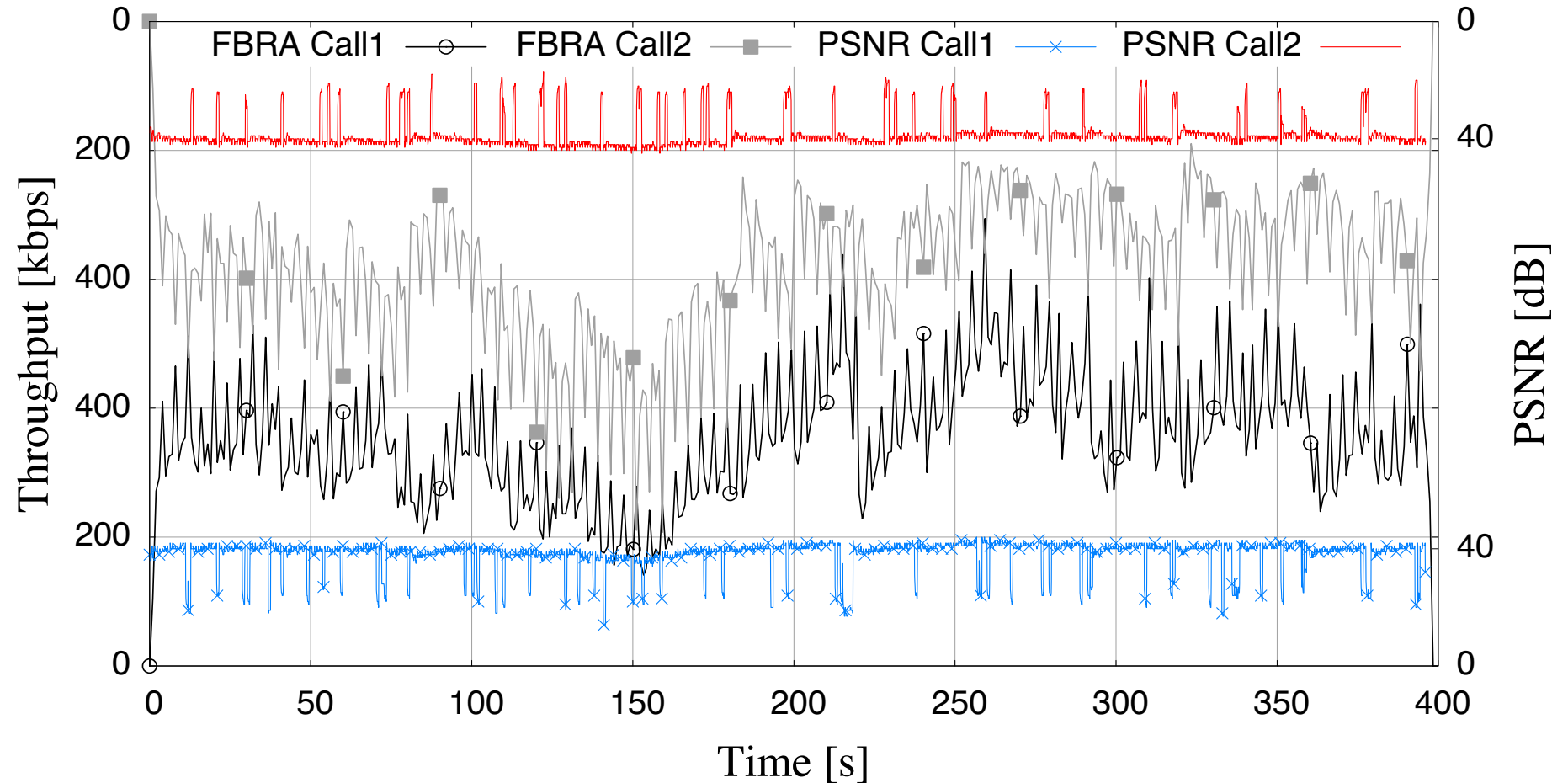


Compete with short TCPs



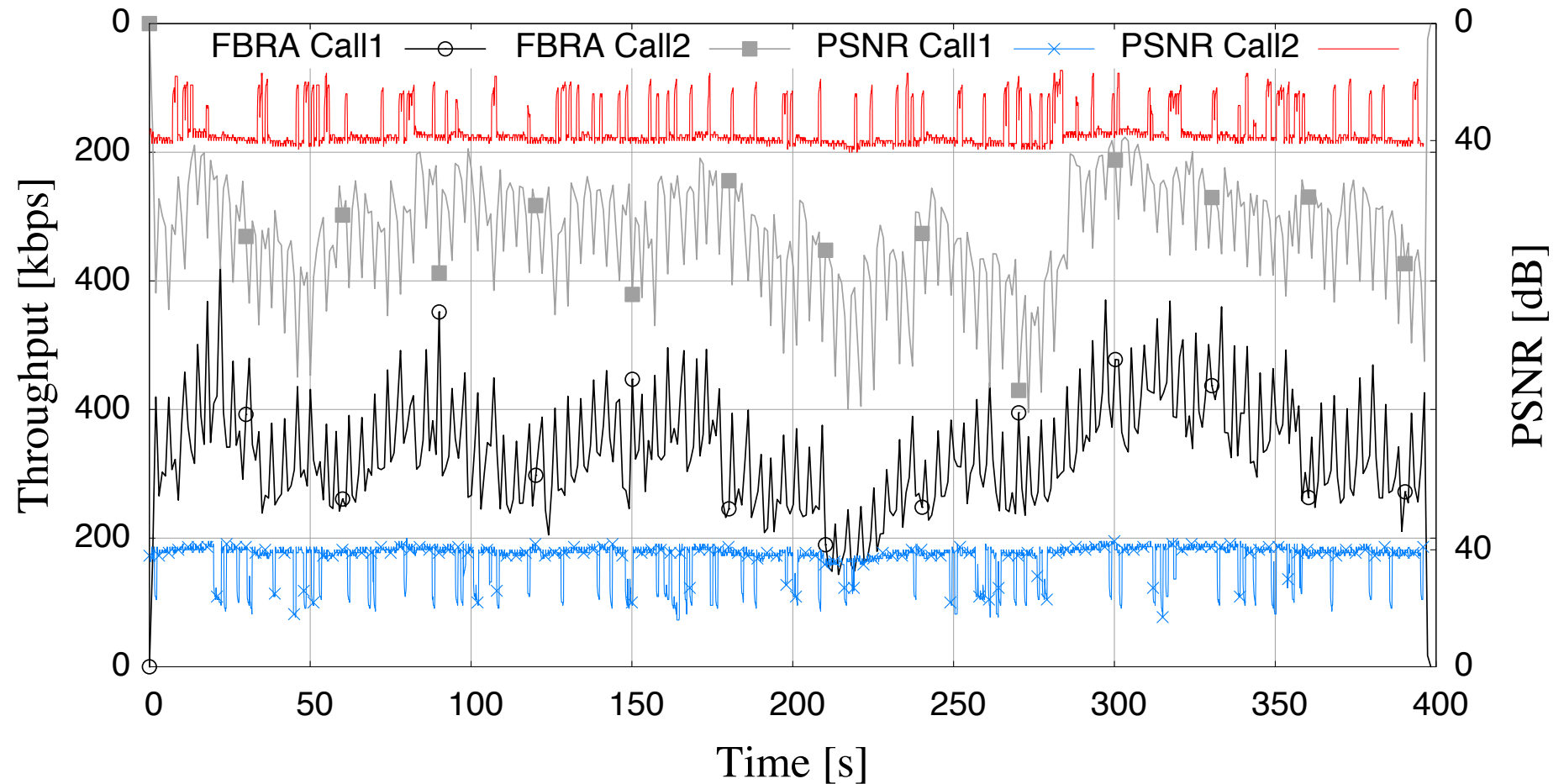
TESTBED Evaluation (1/2)

Two RTP flows on the link



- 1Mbit/s link capacity, 50ms one-way-delay,

TESTBED Evaluation (2/2)



- 1Mbit/s link capacity, 100ms one-way-delay,

Next steps

- Code: (coming soon)
 - <https://github.com/protocols-comnet/rmcat-adaptive-fec-code>
- Evaluation Paper:
 - Nagy M., Singh V., Ott J., Eggert L., Congestion Control using FEC for Conversational Multimedia Communication, Proc. of ACM Multimedia Systems, Singapore, SG, Mar, 2014,
- More feedback is appreciated 😊