Delay and Delay Variation Simulation Results for Multi-hop Conventional Ethernet Cases with Bursting/Bunching

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Outline

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□Add additional cases as needed

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Introduction

Reference [1] presented initial simulation results for transport of time-sensitive traffic over conventional Ethernet

Considered simple, one-hop, two-switch network

• one switch-to-switch link with end-devices attached to each switch

Results showed that unfiltered peak-to-peak delay variation for competing Constant Bit Rate (CBR) traffic streams whose rates differ slightly from nominal can be appreciable compared to the requirements for digital audio and video [2]

 For 3 CBR streams, 50% link utilization, and 256 byte packets, unfiltered delay variation was almost 50 μs in one case

• For sufficiently small frequency offsets (e.g., 1 ppm or less), phase-locked loop (PLL) filtering at the egress did not reduce the delay variation appreciably

- This exceeds the requirements for uncompressed digital video and digital audio, and is close to the limit for compressed digital video (50 µs) [2]
- \bullet For 6 CBR streams, 50 % link utilization, and 256 byte packets, unfiltered delay variation exceeded 100 μs in one case, and was reduced to just over 80 μs by filtering
 - This exceeds the limits for digital audio and video
- Also considered adding a best-effort stream with maximum size packets, though this did not appreciably change the time-sensitive stream results

Introduction (Cont.)

Discussion during the presentation of [1] indicated it would be of interest to show a worse case, with multiple hops

□Further discussion in a subsequent ResE SG conference call indicated it would be of interest for the multiple hop case to resemble the bursting/bunching scenarios described in [3] (see Annex F of [3] for details)

- •Combine *N* traffic sources from *N* locally-attached end devices at a switch, and transport over a link to a downstream switch
- Replicate this configuration N times, so that the downstream switch has N incoming links
- Drop the traffic from N-1 of the sources from each incoming link at the downstream switch to locally-attached end devices (the number of locally attached end devices is therefore N(N-1))
- Transport the remaining N streams (one from each incoming link) over an outgoing link to a downstream switch
- Repeat the above scenario at the downstream switch, i.e., replicate the above configuration N times
- For k stages, the total number of sources at the ingress grows like N^k

Introduction (Cont.)

The analysis in [3] is mainly qualitative, i.e., is carried out by graphically representing packets at various times

- The analysis assumes worst-case arrival patterns (i.e., packets from competing time-sensitive streams always arrive simultaneously)
- The analysis in [3] considers both queueing at the input of each switch and queueing at the output of each switch (these are separate cases)
- □ It was felt it would be desirable to simulate this scenario as a case that is possibly worse than those considered in [1]

□In addition, it is of interest to consider total delay for multi-hop cases

- •While total delay was not explicitly discussed in [1], end-to-end delays for a path consisting of the ingress link, single switch-to-switch link, and egress link were on the order of at most 300 µs, and were this large only for the case that included a single best-effort stream with maximum size packets
 - The longest path through the network was between 100 and 200 m, and the propagation delay was 1.755×10^8 m/s (the default minimum propagation speed in Opnet, which assumes dispersion representative of the medium and configuration
 - With this assumption, propagation delay is of the order a few μ s, and is therefore negligible (and would be negligible with no dispersion)

Simulation Models and Assumptions

□As in [1], OPNET simulation tool was used to simulate packet delays

- •OPNET contains models for full-duplex Ethernet MAC and for Ethernet bridges
- Models were modified (as indicated in [1]) to include priority classes
 - Priority queueing is non-preemptive

Considered basic topology as described in [3] and summarized in Introduction

- •At each stage, combine *N* previous stages
 - Each previous stage supplies *N* traffic streams to this stage
 - Drop the traffic from N 1 of the traffic streams from each previous stage
 - -Therefore, need N(N-1) end devices at this stage
 - Carry one traffic stream from each incoming link (from each previous stage) over an egress link to the next stage
 - -Therefore, *N* traffic streams are carried to the next stage

Assume all packets are maximum size

1500 bytes (Opnet adds Ethernet overhead)

□Assume 100 Mbit/s Ethernet links

□Assume the maximum path length through the network is 100 - 200 m, and the propagation speed is as in [1], i.e., 1.755×10^8 m/s

Simulation Models and Assumptions (Cont.)

Assume all time sensitive traffic streams have the same nominal rate, with a small frequency offset

- Frequency offset is different for each competing stream, and is chosen on input
 - This captures the fact that Time-sensitive video and audio clients have specified nominal rates, but are allowed to differ from those nominal rates by specified frequency tolerances
- Nominal rate is chosen based on the number of streams per switch at the network ingress and desired link utilization
 - Input rate (and time between packets) is constant

□OPNET model assumptions (same as in [1])

- Two priority classes
 - Time-sensitive traffic gets high priority
 - Best-effort traffic gets low priority
 - Priority queueing is non-preemptive
 - Queueing is first-come, first-served (FCFS) within each priority class
- •OPNET model for full-duplex Ethernet MAC is used (with priorities added)
- •OPNET contains spanning tree and rapid spanning tree algorithms
 - Use rapid spanning tree algorithm here

Simulation Case 1

Three sources at the ingress of each switch at the initial stage (N = 3)

□4 stages (k = 4), i.e., a traffic stream that is not dropped at an intermediate stage traverses 4 switch-to-switch links (plus one ingress and one egress link)

- Total of 81 traffic sources
 - We needed to restrict the numbers of hops and/or sources/switch to keep the total number of sources manageable (e.g., N = 3 and k = 7 would have produced 2187 traffic sources; N = 6 and k = 7 would have produced 279936 traffic sources.)

□All traffic is time-sensitive

- Packet size is as given above (1500 bytes plus Ethernet overhead added by Opnet)
- Nominal packet arrival rate for each stream is 1333.33 packets/s
 - Nominal time between packets is 0.00075 s
 - Resulting switch-to-switch link utilization is approximately 50% (results from 3 traffic streams)
 - -Utilization per stream assuming nominal arrival rate and excluding Ethernet overhead is 16%

□Network topology is shown on the next 3 slides

It was convenient to use the subnet capability of Opnet, due to the large number of traffic sources and hierarchical structure of the network

Simulation Case 1 (Cont.)

Simulate for 255 s, with traffic turned on at 5 s

 Needed to add small amount of best-effort traffic in reverse direction to ensure each destination node would be in the forwarding database of each switch (otherwise get flooding and link utilizations that exceed 100%)

•Node_0 sink in stages 3 and 4 is used for some of this reverse traffic

Simulation Case 1 - Stages 1 and 2





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Simulation Case 1 - Stage 3



Switch to Switch link utilization = 50%

Each subnet (the red octagonal icons labeled 9 CBR Sources_1, _2, and _3 represent a stage 1 and 2 network as shown on the previous slide

Simulation Case 1 - Stage 4



Switch to Switch link utilization = 50%

Each subnet (the red octagonal icons labeled 27 CBR Sources_1, _2, and _3 represent a stage 3 network as shown on the previous slide

Simulation Case 1 Traffic Streams - Stages 1 and 2

□Node 2 to node 14, rate offset by -100 ppm

□Node 3 to node 15, rate offset by +100 ppm

□Node 5 to node 16, rate offset by -50 ppm

□Node 6 to node 17, rate offset by +50 ppm

□Node 8 to node 18, rate offset by -75 ppm

□Node 9 to node 19, rate offset by +75 ppm

□Streams from nodes 1, 4, and 7 are transported on link to stage 3 (see following slides for details)

Simulation Case 1 Traffic Streams - Stage 3

□27 CBR Sources_1 subnet

- ■9 CBR Sources_1 subnet, node 4 to node 1, rate offset by -10 ppm
- ■9 CBR Sources_1 subnet, node 7 to node 2, rate offset by +10 ppm
- •9 CBR Sources_2 subnet, node 4 to node 3, nominal rate
- ■9 CBR Sources_2 subnet, node 7 to node 4, rate offset by +10 ppm
- ■9 CBR Sources_3 subnet, node 4 to node 5, rate offset by –10 ppm
- •9 CBR Sources_3 subnet, node 7 to node 6, nominal rate

□27 CBR Sources_2 subnet

- ■9 CBR Sources_1 subnet, node 4 to node 1, rate offset by -10 ppm
- ■9 CBR Sources_1 subnet, node 7 to node 2, rate offset by +10 ppm
- •9 CBR Sources_2 subnet, node 4 to node 3, nominal rate
- ■9 CBR Sources_2 subnet, node 7 to node 4, rate offset by +10 ppm
- ■9 CBR Sources_3 subnet, node 4 to node 5, rate offset by –10 ppm
- •9 CBR Sources_3 subnet, node 7 to node 6, nominal rate

Simulation Case 1 Traffic Streams - Stage 3 (Cont.)

□27 CBR Sources_3 subnet

- ■9 CBR Sources_1 subnet, node 4 to node 1, rate offset by -10 ppm
- ■9 CBR Sources_1 subnet, node 7 to node 2, rate offset by +10 ppm
- •9 CBR Sources_2 subnet, node 4 to node 3, nominal rate
- •9 CBR Sources_2 subnet, node 7 to node 4, nominal rate
- ■9 CBR Sources_3 subnet, node 4 to node 5, rate offset by –10 ppm
- •9 CBR Sources_3 subnet, node 7 to node 6, nominal rate

□Note that the 3rd and 4th streams are different in all three 27 CBR subnets

Simulation Case 1 Traffic Streams - Stage 4

- □27 CBR Sources_1, 9 CBR Sources_1 subnet, node 1 to node 7, nominal rate
- □27 CBR Sources_1, 9 CBR Sources_2 subnet, node 1 to node 1, rate offset by –10 ppm
- 27 CBR Sources_1, 9 CBR Sources_3 subnet, node 1 to node 2, rate offset by +10 ppm
- □27 CBR Sources_2, 9 CBR Sources_1 subnet, node 1 to node 8, rate offset by –10 ppm
- 27 CBR Sources_2, 9 CBR Sources_2 subnet, node 1 to node 3, rate offset by +10 ppm
- □27 CBR Sources_2, 9 CBR Sources_3 subnet, node 1 to node 4, nominal rate
- □27 CBR Sources_3, 9 CBR Sources_1 subnet, node 1 to node 9, rate offset by +10 ppm
- □27 CBR Sources_3, 9 CBR Sources_2 subnet, node 1 to node 5, rate offset by –10 ppm
- □27 CBR Sources_3, 9 CBR Sources_3 subnet, node 1 to node 6, nominal rate













27 CBR Sources_2, 9 CBR Sources_1 subnet, node 4 to node 1, rate offset by -10 ppm

27 CBR Sources_2, 9 CBR Sources_3 subnet, node 7 to node 6, nominal rate

27 CBR Sources_3, 9 CBR Sources_1 subnet, node 4 to node 1, rate offset by -10 ppm

27 CBR Sources_3, 9 CBR Sources_3 subnet, node 7 to node 6, nominal rate









Summary of Peak Delay and Peak-to-Peak Delay Variation

Number of Hops	Peak Delay (ms)	Peak-to-Peak Delay Variation (μs)
1	0.62	250
2	1.0	500
3	1.5	510
4	1.6	850

Conclusions

Peak delay and peak-to-peak delay variation increase with number of hops, as expected

- Peak delay reaches 1.6 ms after 4 hops
- Peak-to-peak delay variation reaches 850 µs after 4 hops

Peak delay is slightly below worst-case that would be obtained for this 4-hop case

- Worst-case for 3 contending CBR streams at a switch occurs when 2 frames are queued when a frame arrives (for link utilization < 100%)
- Then, for contention occurring at 4 switches (in a 4-hop path), the delay due to contention is (4)(2)(frame transmission delay) = 8(frame transmission delay)
- Also have transmission delay for the frame itself on the 4 switch-to-switch links plus the two access links
- Then total delay due to transmission and queueing, in worst case, is 14(frame transmission delay)
- •Then worst-case delay (neglecting propagation delay since it is much smaller) is

• 14(12000+8(38) bits)/10⁸ bits/s) = 1.72 × 10⁻³ s = 1.72 ms

If number of contending traffic streams is increased from 3 to 6 (still for a 4-hop case), would expect worst-case delay due to contention to increase by (4)(3)(frame transmission delay) = 12(12000+8(38)) bits)/10⁸ bits/s) = 1.48 ms

Worst-case total delay in this case would be 1.72+1.48 ms = 3.2 ms

- If number of hops is increased from 4 to 7 (still for 3 contending traffic streams), would expect worst-case delay to increase to [7(2)+9](frame transmission delay] (i.e., 2 contending frames at each of 7 switches plus 9 total transmission delays (switch-to-switch plus access links))
 - •Worst case total delay in this case would be $23(12000+8(38) \text{ bits})/10^8 \text{ bits/s}) = 2.83 \times 10^{-3} \text{ s} = 2.83 \text{ ms}$

Peak-to-peak delay variation is slightly below worst-case that would be obtained for this 4-hop case

- Worst-case peak-to-peak delay variation is equal to the worst-case delay due to contention at the switches, as this is the component of delay that is not always present
- Then worst-case peak-to-peak delay variation is

• 8(12000+8(38) bits)/10⁸ bits/s) = 9.8×10^{-6} s = 980 µs

- □Note that the amount by which the worst case peak delay exceeds the actual peak delay (1.72 ms 1.6 ms = 0.12 ms) and the amount by which the worst case peak-to-peak delay variation exceeds the actual peak-to-peak delay variation (980 μ s 850 μ s = 130 μ s = 0.13 ms) are approximately equal, as expected
- Results obtained for 1, 2, and 3 hops are consistent with similar worst-case analyses for these cases (with 3-hop results below worst-case results by approximately 0.1 ms)
- □While peak delay does not exceed 2 ms for 3 contending traffic streams and 4 hops, the results indicate that it will exceed 2 ms for 6 contending traffic streams with 4 hops, and 3 contending traffic streams with 7 hops

□Worst-case peak-to-peak delay variation is just below 1 ms (i.e., 980 µs) for 3 contending traffic streams with 4 hops

- For 6 contending streams with 4 hops, this increases to 5(4)(frame transmission delay) = 2.46 ms
- For 4 contending streams with 7 hops, this increases to 3(7)(frame transmission delay) = 2.58 ms

□Therefore, while peak-to-peak delay does not exceed 2 ms for 3 contending traffic streams and 4 hops, the results indicate that it will exceed 2 ms for 6 contending traffic streams with 4 hops, and 3 contending traffic streams with 7 hops

□The results indicate the following rules of thumb may be used to estimate worst-case delay and worst-case peak-to-peak delay variation for an arbitrary *N* hop path through a network

•Worst case end-to-end delay = $[(N+2) + \sum_{j=1 \text{ to number of switches}} (\{\text{number of incoming links at switch } j \} - 1)][\text{frame transmission delay}]$

•Assumes propagation delay is negligible (must be added if it is not negligible)

•Worst case peak-to-peak delay variation = $[\Sigma_{j=1 \text{ to number of switches}}(\{\text{number of incoming links at switch } j \} - 1)][\text{frame transmission delay}]$

References

- Geoffrey M. Garner and Felix Feng, Delay Variation Simulation Results for Transport of Time-Sensitive Traffic over Conventional Ethernet, Samsung presentation at July, 2005 IEEE 802.3 ResE SG meeting, San Francisco, CA, July 18, 2005.
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