

An Inline Measurement Method for Capacity of End-to-end Network Path

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Abstract

We previously proposed a new version of TCP, called Inline measurement TCP (ImTCP), in [1]. The ImTCP sender adjusts the transmission intervals of data packets and then utilizes the arrival intervals of ACK packets for available bandwidth estimation. This type of active measurement is preferred because the obtained results are as accurate as those of other conventional types of active measurement, even though no extra probe traffic is injected onto the network. In the present research, we combine a new capacity measurement function with ImTCP in order to enable simultaneous measurement of both capacity and available bandwidth in ImTCP. The capacity measurement algorithm is a new packet-pair-based measurement technique that utilizes the estimated available bandwidth values for capacity calculation. This new algorithm promises faster measurement than current packet-pair-based measurement algorithms for various situations and works well for high-load networks, in which current algorithms do not work properly. Moreover, the new algorithm provides a confidence interval for the measurement result.

Keywords

end-to-end measurement, available bandwidth, capacity, inline measurement, packet pair

1. Introduction

The *capacity* of an end-to-end network path, which is considered to be the smallest capacity of network links along a path, is the maximum possible throughput that the network path can provide. Traffic may reach this maximum throughput when there is no other traffic along the path. The *available bandwidth* indicates the unused bandwidth of a network path, which is the maximum throughput that newly injected traffic may reach without affecting the existing traffic. The two bandwidth-related values are obviously important with respect to adaptive control of the network. In addition, these two values are often both required at the same time. For example, although network transport protocols should optimize link utilization according to capacity, congestion should be also avoided through the use of available bandwidth information. For route selection or server selection in service overlay networks, information concerning both capacity and available bandwidth offers a better selection than either capacity or available bandwidth information alone. For example, when

the available bandwidth fluctuates frequently and the transmission time is long, the capacity information may be a better criterion for the selection. However, when the available bandwidth appears to be steady during the transmission, the available bandwidth should be used for the selection. Moreover, the billing policy of the Internet service provider is based on both the capacity and the available bandwidth of the access link that they are providing to the customer.

Several passive and active measurement approaches exist for capacity or available bandwidth. Although active approaches are preferred because of their accuracy and measurement speed, sending extra traffic onto the network is a disadvantage that is common to all active measurement tools. For example, Pathload [2] generates between 2.5 and 10 MB of probe traffic per measurement. The average per-measurement probe traffic generated by Spruce [3] is 300 KB. For routing in overlay networks, or adaptive control in transport protocols, these measurements may be repeated continuously and simultaneously from numerous end hosts. In such cases, the probes will create a large amount of traffic that may degrade the transmission of other data on the network, as well as the measurement accuracy itself.

We therefore propose an active measurement method that does not add probe traffic to the network. The proposed method uses the concept of “plugging” the new measurement mechanism into an active TCP connection (*inline measurement*). We previously introduced ImTCP (Inline measurement TCP) [1], a Reno-based TCP that deploys inline measurement. The ImTCP sender not only observes the ACK packet arrival intervals in the same manner as TCP Westwood [4], but also actively adjusts the transmission interval of data packets, in the same way that active measurement tools use probe packets. When the corresponding ACK packets return, the sender utilizes the arrival intervals to calculate the measurement values.

The available bandwidth measurement algorithm for ImTCP is described in detail in [5]. For each measurement, the ImTCP sender searches for the available bandwidth only within a given search range. The search range is a range of bandwidth that is expected to include the current available bandwidth and is calculated statistically from the previous measurement results. By introducing the search range, sending packets at an extremely high rate can be avoided. This also allows the number of packets for the measurement to be kept small, so that measurement is still possible when the TCP window size is relatively small. The search range is divided into multiple sub-ranges of identical width of bandwidth. For each of the sub-ranges of the bandwidth, the sender transmits a group of TCP data packets (a packet stream), the transmission rate of which varies to cover the sub-range. The sender then determines whether an increasing trend exists in the transmission delay of packets in each stream when the echoed (ACK) packets arrive at the sender host. Delayed ACKs is supposed to be disabled at the receiver. The increasing trend indicates that the transmission rate of the stream is larger than the current available bandwidth of the network path [2]. This fact allows the sender to infer the location of the available bandwidth in the search range. The simulation results show that the ImTCP sender can perform periodic measurements at short intervals, on the order of several RTTs

and the measurements results reflect well the changes in the available bandwidth of the network.

In the present paper, we introduce an inline measurement algorithm for capacity for ImTCP. The proposed algorithm utilizes the arrival intervals of the ACK packets of packet pairs (PPs) that are sent back-to-back. In the existing PP-based capacity measurement algorithm [6-8], the PPs that are cut into by other packets from cross traffic at the bottleneck link causes incorrect capacity estimation and are therefore eliminated from the data used in the calculation. Unlike previous algorithms, the proposed algorithm in ImTCP can use these PPs for capacity measurement, which enables ImTCP to collect more information from PPs so that faster and more accurate measurement can be expected.

The main concept of the proposed capacity measurement algorithm of ImTCP is that the available bandwidth information, which can be yielded periodically due to the deployed available bandwidth measurement mechanism, is exploited. The available bandwidth information is used for estimation of the quantity of the cross traffic that cuts in PPs at the bottleneck link so that the interval of the PPs becomes usable for the capacity measurements. The proposed algorithm also uses statistic analysis to calculate the confidence interval of the delivered results.

Through simulation validations, we show that ImTCP can deliver measurement results quickly, independent of the characteristics of the network. In addition, we find that the capacity measurement algorithm works well in extremely high-load networks, in which current measurement algorithms do not work well.

The remainder of this paper is organized as follows. In Section 2 we discuss PP-based measurement techniques used for inline measurement. In Section 3, we introduce the proposed measurement algorithm for network capacity. In Section 4, we evaluate its performance through simulation experiments. Finally, in Section 5, we present concluding remarks and discuss future projects.

2. Packet-pair-based capacity measurement algorithm

Currently there are various approaches for measuring the capacity of an end-to-end network path [9-13]. Some of these approaches use packets of various size to probe the network and infer the network capacity from the difference in the transmission delays of packets of various sizes [9]. Other approaches use the probe packets in different TTLs (Time To Live) to measure all link bandwidth, rather than just the capacity of the bottleneck link [10-13]. These approaches can not be used for inline measurement because changes in TCP data packet size or TTLs for the purpose of measurement may cause severe deterioration in the data transmission throughput of TCP. We found that only packet-pair-based measurement can be used for inline measurement because no changes in packet size or TTL are required, whereas packets that are sent back-to-back can be created with the current ImTCP structure without requiring any changes.

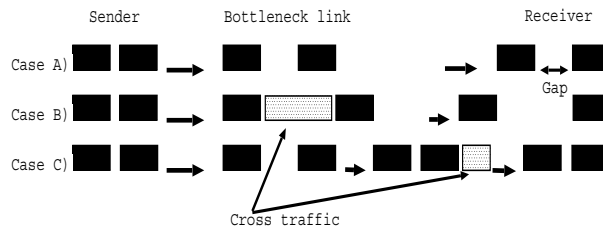


Figure 1: Three cases showing how the spacing between a pair of packets may change as the pair travels along a path.

2.1 Packet pair technique

The intuitive rationale of capacity measurement using packet pairs is that if two packets are sent close enough together in time to cause the packets to queue back-to-back at the bottleneck link, then the packets will arrive at the destination with the same spacing as when they left the bottleneck link [9]. The spacing is supposed to remain unchanged until the packet pairs reach the receiver, as shown in Case A of Figure 1, which is a variation of a figure taken from [7]. In this case, the capacity of the bottleneck link (C) can be calculated by the equation:

$$C = \frac{P}{Gap} \quad (1)$$

where P is the size of the packet pairs, and Gap is the time spacing of the two packets when arriving at the receiver.

However, when a packet pair travels along the path, two more situations can occur. As shown by Case B in Figure 1, the two packets may be cut into by other packets from cross traffic at the bottleneck link. The result is that, the spacing between the two packets becomes larger than expected. In this case, Equation (1) leads to an under-estimation of the capacity. In another case, indicated by Case C in Figure 1, the packet pairs may pass back-to-back through the bottleneck link, but in a link downstream of the bottleneck link, the pairs again get in queue, and the spacing between the two packets is shortened. In this case, Equation (1) leads to over-estimation.

Current PP-based measurement techniques use only the PPs described in Case A to calculate capacity. These techniques have various mechanisms for determining the Case-A PPs from all of the received PPs. Some tools assume a high frequency of appearance of Case-A PPs and so search for these PPs from a frequency histogram (Pathrate [6]) or a weighting function (Nettimer [7]). CapProbe [8] repeatedly sends packet pairs until it discovers a Case-A PP, based on the transmission delay of the packets.

When the network path is almost empty, Case-A PPs may appear with the highest frequency. However, when other traffic appears on the network, the bottleneck link is often congested, and so the probability of a Case-B PP appearing is much higher than that of a Case-A or Case-C PP. In this case, CapProbe will spend an extremely long time for capacity searching, and Pathrate and Nettimer will deliver incorrect estimations.

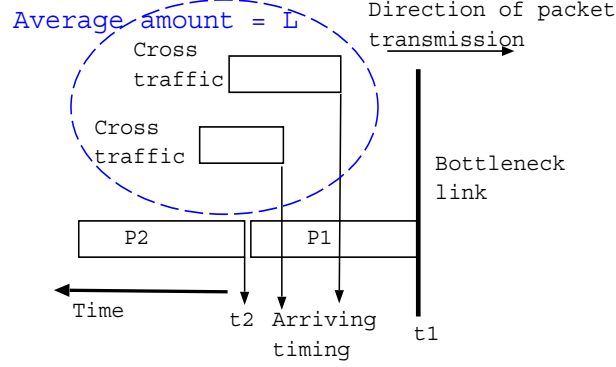


Figure 2: Arrival time at the bottleneck link of packet pairs and cross traffic

Unlike those existing techniques, we propose a new technique by which to calculate capacity that can use both Case-A PPs and Case-B PPs. This is possible because of the available bandwidth information that is available in ImTCP.

2.2 Capacity calculation

Let us consider the timing of the arrival at the bottleneck link of a PP (Figure 2). We assume that the first packet arrives at t_1 and the second packet arrives at t_2 . During the interval from t_1 to t_2 , packets from other traffic may arrive at the bottleneck link. The second packet (P2) must wait in the queue for the processing of the cross packets. Therefore, the time spacing (*Gap*) of the packet pair after leaving the bottleneck link is the total of the queuing time and the processing time of the second packet. That is:

$$Gap = \frac{P + L}{C} \quad (2)$$

where L is the amount of the cross traffic that arrives at the bottleneck link during the interval (t_1, t_2) . Supposing that the bottleneck link of a network path is the link having the smallest available bandwidth, we can then calculate the total transmission rate of the cross traffic at the bottleneck link as: $C - A$, where A is the current available bandwidth. Let δ be the time spacing of the packet pair upon arrival at the bottleneck link ($\delta = t_2 - t_1$). Then, the average value of L is:

$$L = \delta(C - A) \quad (3)$$

from Equation (2) and (3), we can write:

$$C = \frac{P + \delta(C - A)}{Gap},$$

or

$$C = \frac{P - \delta \cdot A}{Gap - \delta}. \quad (4)$$

Equation (4) enables the calculation of capacity from the PPs for both Case A and Case B. In the next section, we propose the new capacity calculation algorithm based on the equation.

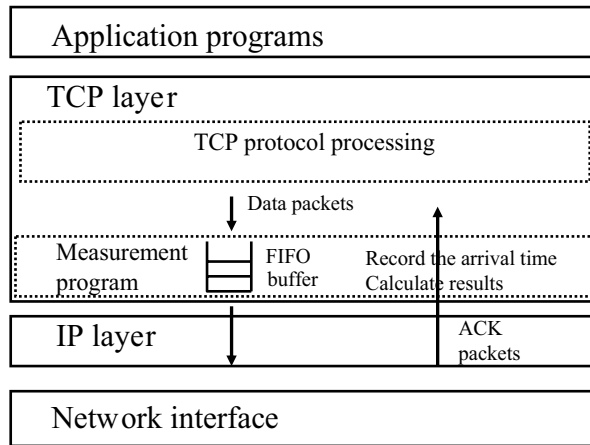


Figure 3: Placement of measurement program at ImTCP sender

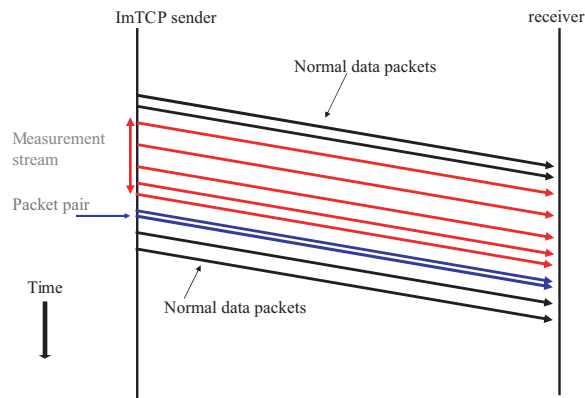


Figure 4: Creation of packet pairs in ImTCP

3. Inline measurement algorithm for capacity

3.1 Implementation of packet pairs in ImTCP

As introduced in our previous study [1], a measurement program is inserted into the sender program of TCP Reno to create an ImTCP sender. The measurement program is located at the bottom of the TCP layer, as shown in Figure 3. When a new data packet is generated at the TCP layer and is ready to be transmitted, the packet is stored in an intermediate FIFO buffer. The measurement program waits until the number of packets in the intermediate buffer becomes sufficient and then decides the time at which to send the packets in the buffer in order to create measurement streams. When no measurement stream is needed, the program immediately passes

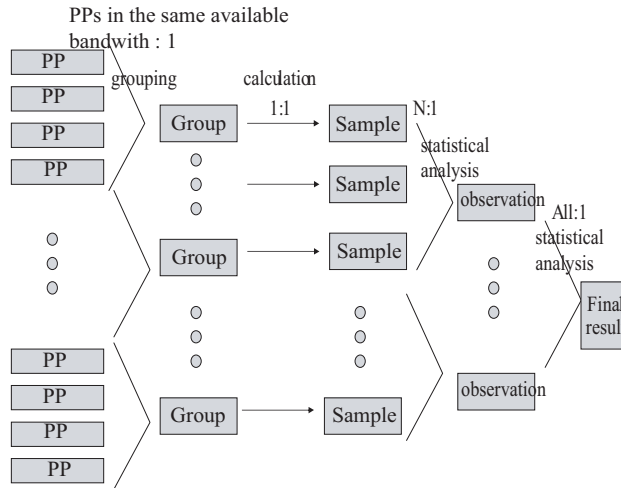


Figure 5: Proposed algorithm

all of the data packets to the IP layer. In the previous version of ImTCP [1], we decided that the program forms and sends one measurement stream for the available bandwidth in each RTT in order to maintain fairness with respect to traditional TCP Reno.

To the current system, we add the creation and transmission of a PP just after the transmission of each measurement stream, as depicted in Figure 4. Note that during the transmission of a measurement stream, some packets may arrive at an intermediate FIFO buffer, so that there are usually a number of packets available in the buffer just after the transmission of a stream [1]. Therefore, there is almost no waiting time required in the creation of a packet pair. Therefore, there is almost no effect on the performance of ImTCP by introducing the capacity measurement algorithm.

In ImTCP, 2–4 measurement streams are required in order to determine the available bandwidth. As mentioned above, each PP is formed and transmitted after each measurement stream. Therefore, 2–4 results for PPs can be obtained during the interval of two consecutive measurement results for available bandwidth.

3.2 Proposed measurement algorithm

We next explain the procedure for determining the capacity from the measurement results of PPs using Figure 5. The procedure involves the following steps:

- **Grouping of PPs:** PPs that sent when the measured available bandwidth remains unchanged are placed in the same group. The average value of arrival interval of PPs in a group, denoted by \overline{Gap} , is then calculated. To obtain a good average value, the number of PPs in each group should be enough large i.e. larger than or equal to 3, as determined herein. Therefore, after grouping, a group having only one or two PPs will be merged with a nearby group.

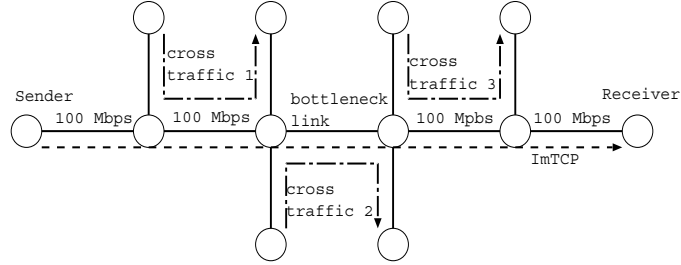


Figure 6: Simulation topology

- Calculation: Based on the \overline{Gap} value of a group, a sample of capacity is calculated using the following function.

$$\text{If } \frac{A}{P/\delta} > \lambda$$

$$C = \frac{P}{\overline{Gap}} \quad (5)$$

otherwise,

$$C = \frac{P - \delta \cdot A}{\overline{Gap} - \delta} \quad (6)$$

where λ is the threshold showing the relation between the available bandwidth and the rate of the PPs upon arriving at the bottleneck link, that is defined as P/δ . We assume that the links before the bottleneck link do not have a noticeable effect on the time space, so that δ is approximated by the time interval in which the sender sends the packets. When the available bandwidth is approximately equivalent to the rate of the PPs upon arriving at the bottleneck link, which is considered as $\frac{A}{P/\delta} > \lambda$, the packets may pass through the link without being cut into by other packets (Case A). In this case, Equation (5) (based on Equation (1)) is used. On the other hand, since when the arrival rate of the PPs is much higher than the available bandwidth, which is considered as $\frac{A}{P/\delta} \leq \lambda$, the probability is high that the PP is a Case-B PP, Equation (6) (based on Equation (4)) is used.

- Statistical analysis:
 - We form observations, each of which is the average value of N samples. N should be large enough so that each observation has high accuracy. But when N is too large, the time required to finish an observation is long. This means that the proposed algorithm can not deliver the measurement results quickly. In the present paper, based on empirical experiments, we recommend $N = 10$.
 - The average value of the observations are calculated as the “final result”. The 90% confidence interval is also calculated to show the degree of fluctuation of the capacity.

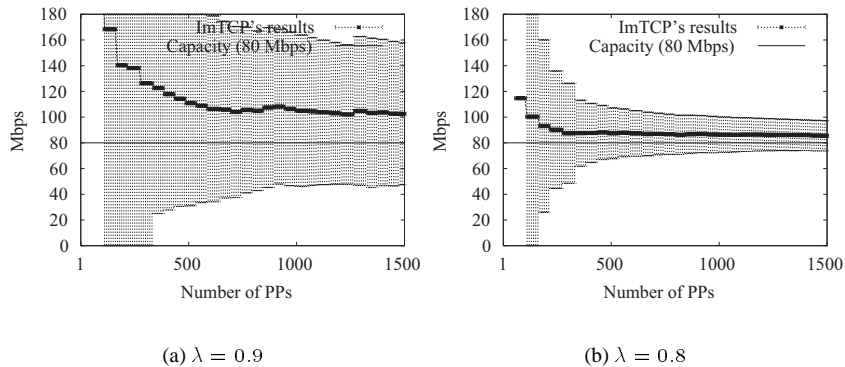


Figure 7: Measurement results for the proposed algorithm when Cross traffic 2 is 5 Mbps. The errors bars show the 90% confidence interval of the correspondent results. $\lambda = 0.8$ gives more accurate results than $\lambda = 0.9$.

4. Simulation experiments

In this Section, we examine the measurement results of the proposed capacity measurement algorithm through ns-2 [14] simulations. We also compare the proposed algorithm with two existing algorithms, CapProbe [8] and Pathrate[6] in the scope of inline measurement.

We use the simulation topology shown in Figure 6. The transmission rate of Cross traffic 1 is fixed to 5 Mbps and that of Cross traffic 3 is fixed to 15 Mbps. The packet size distribution of cross traffic is set to the statistical results for the Internet traffic reported in [15]. This mixture has an average packet size of 404.5 bytes and has a correlation value of 0.999 when compared to realistic Internet traffic. The simulation time is 80 (s).

4.1 Effect of parameters

- Value of λ

We set the bottleneck link capacity to 90 Mbps and the transmission rate of Cross traffic 2 to 5 Mbps and examine the measurement results when $\lambda = 0.9$ (Figure 7(a)) and $\lambda = 0.8$ (Figure 7(b)). These figures show the changes of the capacity measurement results as the number of PPs sent for the measurement increases. The errors bars show the 90% confidence interval of the correspondent results. For the first some PPs, there is at most 1 “observation” is delivered therefore ImTCP can not calculate the confidence interval. In this case, the load on the bottleneck link is low, so the Equation (5) should normally be used. The setting $\lambda = 0.9$ does not allow the Equation (5) to be used so frequently and therefore leads to a bad result, that can be seen in large confidence intervals. We see that in this case $\lambda = 0.8$ or (lower than 0.8) is a better setting.

We next show the case when the capacity is 80 Mbps and the rate of Cross traffic 2

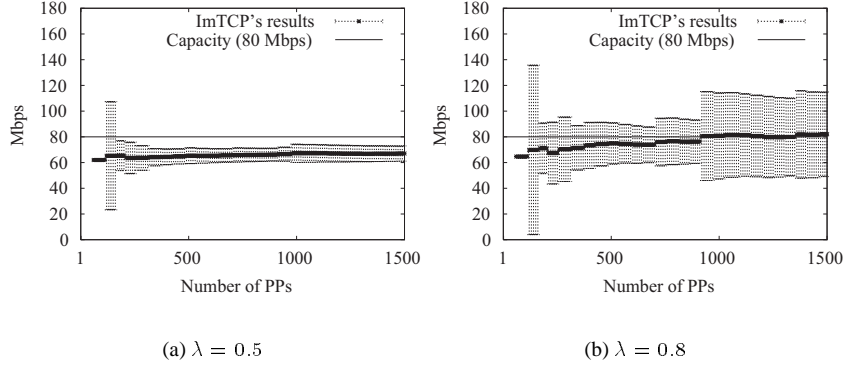


Figure 8: Measurement results for the proposed algorithm when Cross traffic 2 is 20 Mbps. $\lambda = 0.8$ gives more accurate results than $\lambda = 0.5$.

is set to 20 Mbps in Figure 8(a) ($\lambda = 0.5$) and 8(b) ($\lambda = 0.8$). In this case, the rate of the cross traffic is high, so Equation (6) should normally be used. Therefore, a small value of λ , such as 0.5, gives incorrect results for the capacity, and, again, $\lambda = 0.8$ is a good setting in this case. Thus, $\lambda = 0.8$ is a suitable setting for the two cases above, and we found that it is a good setting in many other cases. Therefore, in the following simulations, we used $\lambda = 0.8$.

We can see that the measurement results of ImTCP are sometimes not exactly the right value (for example results in Figure 7(b)). The reason for this is that, we suppose that the amount of the traffic that cut in every PPs is the average value of that ($L = \delta(C - A)$), but the amount of traffic that cut in a certain PP is sometimes too large or too small in comparison with L . In these cases, the Sample calculated from these outstanding values (using Equation (6)) is far from the right value of Capacity, this leads to a slight inaccuracy in the final result of ImTCP. However, for longer connections, because the larger number of PPs is sent, ImTCP's results approach nearer to the right value.

- Value of N

N is the number of samples to form an observation. We set the bottleneck link capacity to 80 Mbps and the rate of Cross traffic 2 to 40 Mbps. Figures 9(a), 9(b) and 9(c) show the measurement results when N is set to 1, 10 and 50, respectively. The large confidence interval in Figure 9(a) indicates that a small value of N ($N = 1$) is not suitable. On the other hand, Figure 9(c) indicates that a value of N ($N = 50$), that is too large, is unsuitable as well, because in this case the time required for the results to become stable is long. Figure 9(b) shows the results with the proposed setting ($N = 10$), which can provide fast and good results.

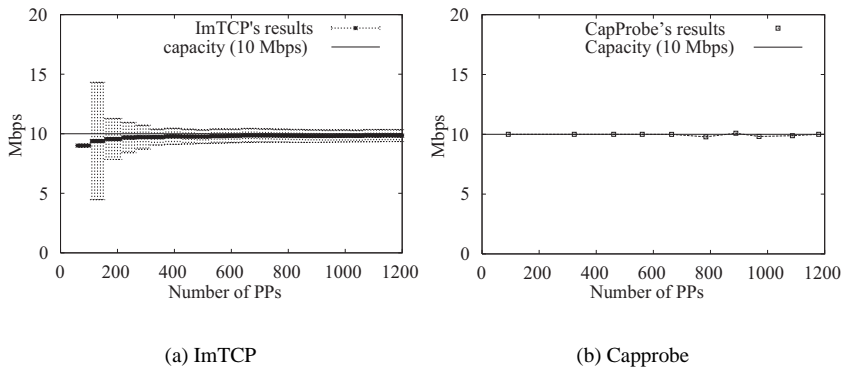


Figure 10: Measurement results for the proposed algorithm (ImTCP) and CapProbe in small capacity, low network load scenario. Both deliver accurate results.

Table 1 Number of PPs required for the first measurement result of the proposed algorithm and CapProbe. CapProbe requires more PPs to deliver the first result.

Capacity	Cross traffic 2	Proposed Algorithm	CapProbe
10 (Mbps)	1 (Mbps)	60 (PPs)	87 (PPs)
10	2	60	85
10	4	60	92
10	5	60	159

advantage of CapProbe is that, compared with the proposed algorithm, CapProbe is simple because it requires no complicated calculations. However, CapProbe only delivers a measurement result after sending a large number of PPs. Table 1 shows the number of PPs sent until the proposed algorithm and CapProbe deliver the first measurement result. Here, the capacity of the bottleneck is kept unchanged while the Cross traffic 2 is varied from 1 Mbps to 5 Mbps. The table shows that, CapProbe only delivers a measurement result after 85 PPs or more are sent. The required number of PP is larger as the network load increases. In contrast, the proposed algorithm delivers good measurement results faster, after sending 60 PPs.

Large capacity, high network load scenario

The capacity is set to 80 Mbps, and the rate of Cross traffic 2 is set to 60 Mbps. In a network with such a heavy load, the proposed algorithm can perform well (Figure 11(a)), whereas CapProbe can not deliver accurate results (Figure 11(b)), because, in this case, most of the PPs are cut into by other traffic so there are very few Case-A PPs. In Figure 11(b) we also show the measurement results of CapProbe when the Cross traffic 2 is decreased to 50 Mbps. These measurement results are still far from the correct value. We believe that CapProbe will perform better if the size of PPs is

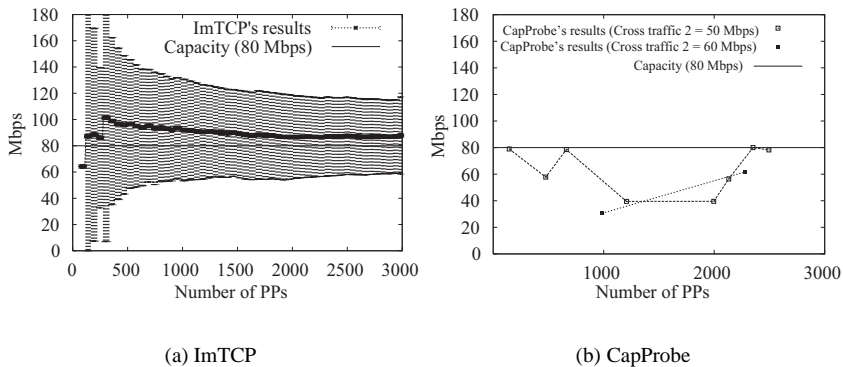


Figure 11: Measurement results for the proposed algorithm (ImTCP) and CapProbe in high network load scenario. CapProbe can not deliver accurate results.

adapted appropriately, instead of being unchanged in the simulations, but changing packet size is not suitable with inline measurement.

4.3 Comparison with Pathrate

In order to accommodate the Pathrate algorithm into TCP, we use the interval of PPs delivered in ImTCP to form the histogram to be used in Pathrate. Pathrate also requires the measurement results of packet trains, referred to as the Average Dispersion Rate (ADR) in the Pathrate algorithm [6]. However, integrating the packet train into TCP is difficult because this has an adverse effect on the performance of TCP. Therefore, we perform the packet train measurement separately from TCP connection, in the same environment as that in the simulation with the ImTCP connection. The result of ADR is then used to find the measurement result for Pathrate.

We use the same network topology as that for the above-described simulations. The capacity is set to 80 Mbps and the transmission rate of Cross traffic 2 is variable. We show the case when the cross traffic contains mainly packets of small size, by randomly varying the packet sizes of the cross traffic within the range of 400 to 600 KB, because in this environment the difference between the proposed algorithm and Pathrate appears clearly. In this case, since most PPs are cut into by cross traffic packets, Pathrate should not work very well. The performance of the proposed program in this environment is also examined, and the measurement results are listed in Table 2.

We explain in detail the respective behaviors of these two algorithms in Figures 12(a) and 12(b). In Figure 12(a), the “Raw data” histogram indicates the measurement results calculated using Equation (1) that are used in Pathrate, and in Figure 12(b), the “Proposed method” histogram shows the “observation” results obtained using proposed algorithm, when the Cross traffic 2’s rate is 75 Mbps. In this case, Pathrate fails to deliver good measurement results because in this case number of

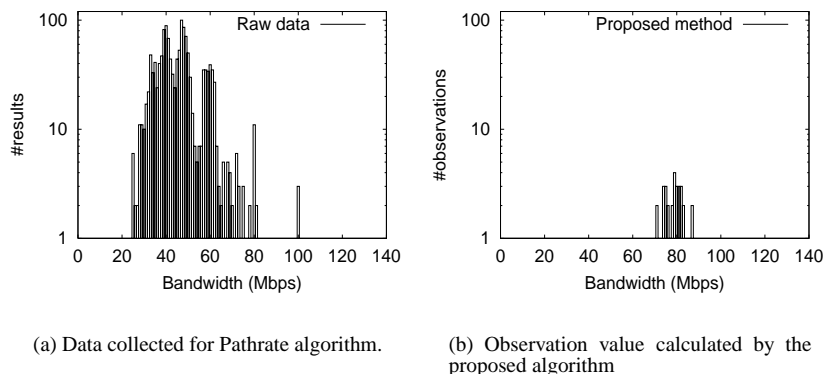


Figure 12: Comparison of the histograms collected by Pathrate and the proposed algorithm in heavy load network (Cross traffic 2’s rate is 75 Mbps).

Table 2 Measurement results of ImTCP and Pathrate when Cross traffic 2 changes. The proposed algorithm can deliver accurate results in high-load network, in which Pathrate does not work well.

Cross traffic 2	ImTCP’s results (90% confidence interval)	Pathrate
75 (Mbps)	79.35 (18.26)	49.00
60	80.24 (23.03)	48.00
40	78.32 (26.04)	80.00
10	81.57 (46.98)	80.00

Case-A PPs are fewer than Case-B PPs. This can be seen in some high peaks near 50 Mbps (while the correct value of capacity is 80 Mbps) in Figure 12(a). In contrast, the proposed algorithm can deliver good results, because the “observation” values always concentrate at the correct value of capacity, regardless of the network load.

5. Conclusion and future works

In this paper, we have proposed a new capacity measurement technique that is suitable for use in TCP connections. In contrast to existing techniques, the proposed mechanism uses available bandwidth information that is available in ImTCP, which enables the utilization of packet pairs that can not be used in existing techniques to calculate the capacity. The simulation results show that, the proposed technique can deliver measurement results quickly, even for a heavily loaded network, in which other techniques do not work well. We are currently implementing ImTCP using the proposed technique on a FreeBSD system. We will also consider a bandwidth measurement algorithm that can be deployed at the TCP receiver.

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