

An Improved RED Algorithm with Input Sensitivity



Kiran Chhabra, Manali Kshirsagar and Arun Zadgaonkar

Abstract Random Early Detection (RED) is a recommended scheme of Active Queue Management (AQM) to avoid network congestion by Internet Engineering Task Force (IETF). RED being sensitive to its control parameters setting and traffic load behaves differently during different phases of scenarios and its performance degrades. To solve the above problems, we have proposed an improved algorithm LTRED, L stands for queue length and T stands for threshold, which incorporates the impact of load variation in early congestion notification along with tuning of threshold parameter of RED. Our approach is validated by doing extensive simulations on ns-2 (network simulator) and comparisons are done with RED, ARED, and AVQ well-known AQMs. Our approach has shown its superiority in terms of packet loss, early congestion indication, better link utilization, and improved goodput. In this work, mismatch between macroscopic and microscopic behavior of queue length of RED is also addressed and reduction in mismatch is achieved. In addition, LTRED makes very few changes to the original RED without making the system complex.

Keywords Random early detection (RED) · Active Queue Management (AQM) IETF (Internet Engineering Task Force) · ns-2 network simulator

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1 Introduction

Due to the gigantic growth of the Internet, it faces the challenge of congestion [1, 2]. AQM is the solution to this congestion control. Its two main objectives are high link utilization with low packet loss and low packet queuing delay. Recommended by IETF [3] RED was first AQM, proposed by Sally Floyd and Jacobson in 1993 [4].

In RED algorithm, Exponential Weighted Moving Average (EWMA) filter as given by Eq. (1) is used for averaging of current queue size. For detection of actual congestion not transient one, averaging procedure is used. Here, two thresholds TH_{\min} minimum threshold and TH_{\max} maximum threshold are used and the average queue size is compared with these thresholds. If the average queue size is less than TH_{\min} no packet is dropped and if it is greater than TH_{\max} every incoming packet is dropped. When it lies between two thresholds, packets are randomly dropped using drop probability calculated by Eqs. (2) and (3).

$$avg_{i+1} \leftarrow (1-w_q)avg_i + w_qq_i \quad (1)$$

$$P_b \leftarrow P_{\max}(avg - TH_{\min}) / (TH_{\max} - TH_{\min}) \quad (2)$$

$$P_a \leftarrow P_b / (1 - count \times P_b), \quad (3)$$

where

avg_{i+1}	average queue size at $(i + 1)$ th time
avg_i	average queue size at i th time
w_q	moving weighted average constant
q_i	current queue size
TH_{\min}	minimum threshold for queue
TH_{\max}	maximum threshold for queue
P_{\max}	maximum value for P_b
P_a	current packet marking probability
P_b	temporary probability used in calculation
count	packets since last discarded packet

Being simple in its approach, RED achieves goal of congestion avoidance, removes global synchronization, and overcomes drawback of Drop tail [3]. Despite having been widely used in combination with TCP for several years, RED has not found acceptance in the Internet research community [1]. It has some disadvantages like hard parameter setting, congestion depends on parameters, insensitivity toward input traffic and there is no matching between average queue length and current queue length [1]. After RED's proposal, lot of research is carried out to find its suitability. Many variants of RED [2, 5–9] were proposed to overcome shortcomings of it with different congestion notification approaches used like the average queue size, packet loss and link utilization and control theory, etc.

The remainder of this paper is organized as follows. Section 2 gives the proposed work along with approach used, Sect. 3 gives simulation results and Sect. 4 deals with conclusions and the final section gives all the references used.

2 Proposed Work

Due to the exponential growth of the Internet, AQM fails to respond fast to traffic changes or to adapt to time-varying TCP dynamics. This motivated us for our proposed work. We have tuned the threshold parameter of RED that is threshold upper limit and lower limit of the average size in terms of router buffer size. Setting TH_{min} to an appropriate value could help router to make good link utilization. Improper setting of TH_{max} also causes more packets losses once the average queue size exceeds it. We have tuned TH_{min} to 40% of maximum queue size and TH_{max} to 70% of it. This tuning has given us good results as well as it helps to achieve the link utilization by properly keeping router buffer efficiently utilized. Since RED's performance degrades heavily shown by wild oscillations when the traffic load becomes bursty indicating it is insensitive to input. We have incorporated this problem in our approach. A little amount work is already done in [10]. LTRED algorithm is shown in Fig. 1.

2.1 Working of Algorithm

For each packet arrival, average value for the queue size is calculated using EWMA as in original RED. To have impact of input traffic the current queue size is compared with a threshold which is 75% value of maximum buffer value (q_{t1}) and depending upon that average value is updated by 5% of the current queue size. This step has impact of input sensitivity. Then it proceeds as RED algorithm. In order to find out if the congestion is sustained once again two checks are made. First, the current queue size is compared with another threshold (85% value of maximum buffer value— q_{t2}) for bursty input and the average queue value is compared with a reference value which is mid value of upper and lower threshold value. If this comparison is found to be true, then packets are dropped earlier as compared to RED algorithm and this is early congestion indication to sources so that they should slow down their rate of sending packets. In rest part, it will continue to work as in original RED.

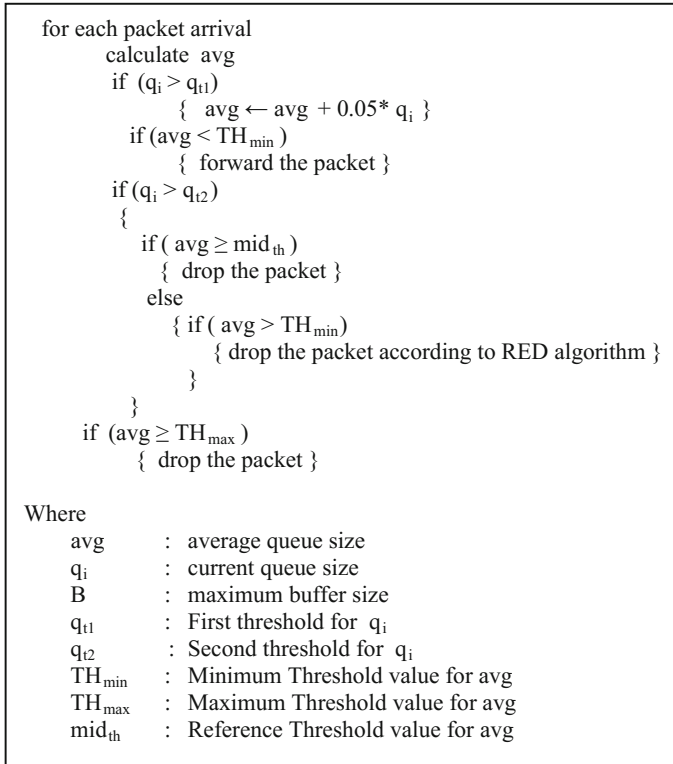


Fig. 1 LTRED algorithm

3 Simulation Results and Discussion

We have implemented the proposed scheme using network simulator ns-2 [11]. To evaluate the improvements network topology is shown in Fig. 2. Bandwidth and delay associated with links are also given in the diagram and Gs to Gr is bottleneck link. We have compared proposed scheme LTRED with RED [4], ARED [7] and AVQ [2], standard AQMs and plotted graphs for various cases.

Total simulation time period is 30 s for the topology shown Two FTP sessions randomly start in between 0 and 0.01 s and lasts till the end. In the middle of simulation, another “m” FTP session would randomly start in between 10.0 and 10.1 s which is to simulate changes of network conditions. TCP Reno is used for all AQMs (RED, ARED, AVQ and LTRED). Parameters used are, for RED and ARED are $TH_{max} = 15$, $TH_{min} = 5$, queue size $q = 30$, $Maxp = 0.1$, $w_q = 0.002$, and for ARED and AVQ other parameters are set as per default value as in ns-2. In case of LTRED $TH_{min} = 12$ and $TH_{max} = 21$, $q_{t1} = 22.5$, $q_{t2} = 25.5$, and $mid_{th} = 16.5$ rest settings are similar to RED. We have observed results for sources varying “m” from 30 to 60 (Tables given) and drawn different graphs for 60 sources

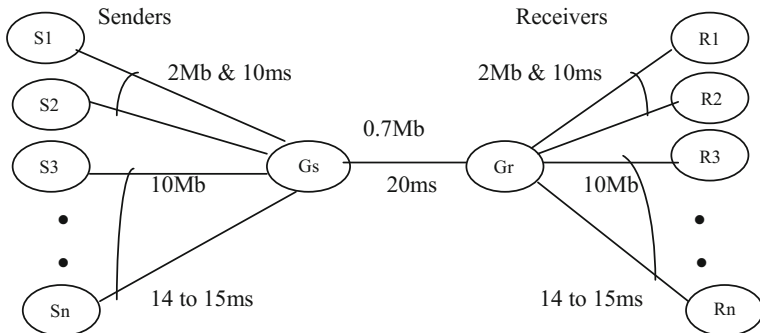


Fig. 2 Simulation network topology

for all the approaches used. We have observed performance in case of packet loss, packet arrival ratio, the average queue size, and the current queue size.

Figure 3 shows the graph for comparison of packet loss for all the four cases showing minor differences during initial period of less traffic. After 10 s when traffic increases that is during congestion period LTRED shows less number of packet losses overall, and shows improvement by 2.3–4.3%. Figure 4 shows total number of packets received for all cases, it shows that packets delivered number is largest in case of LTRED, showing improvement in throughput by 2.19–3.48%. In Fig. 5, the graph shows effective congestion notification to sources in terms of packet arrival rate. Sources are earlier informed about congestion and they reduce their sending rate here also LTRED outperforms other AQMs.

Figure 6 depicts current queue changes, showing more variation in transient congestion and later on in highly congestion area, AVQ uses less buffer size Wild oscillations in RED and ARED show unstable behavior and in case of LTRED nearly stable behavior. Figure 7 illustrates bandwidth utilization which shows that

Fig. 3 Packet loss ratio

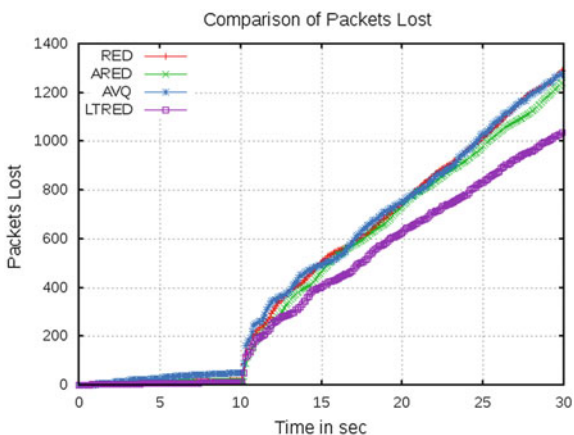


Fig. 4 Packet delivery ratio

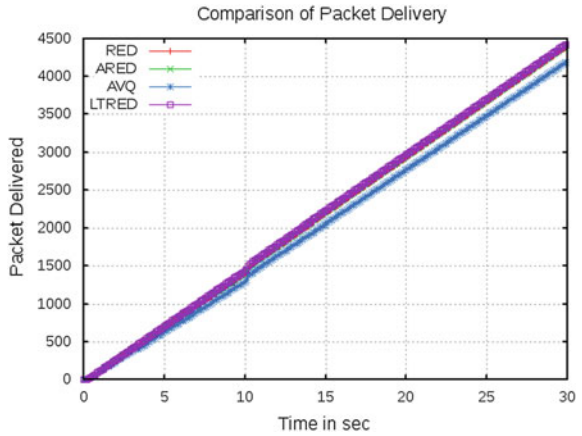
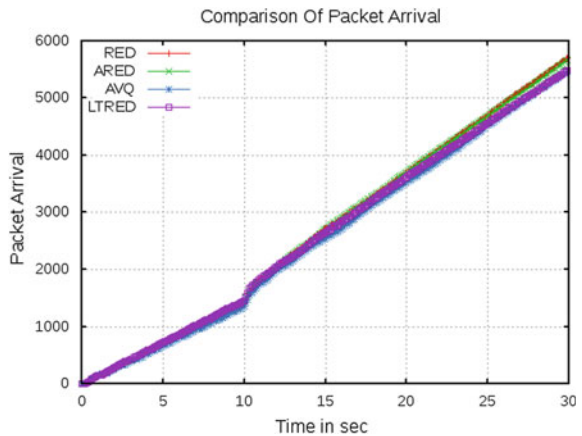


Fig. 5 Packet arrival ratio



LTRED outperforms other AQMs. Figure 8 shows average queue variation for RED, ARED, and LTRED, EWMA approach is used in these cases to calculate the average queue size. In this case also LTRED outperforms RED and ARED by being within its reference limit and depicts stable behavior. Figures 9, 10, and 11 show comparison of the average and the current queue size for RED, ARED, and LTRED cases. Out of these in LTRED case, mismatch behavior of the average and the current queue size is reduced a lot. Tables 1, 2, 3, and 4 give simulation results for number of sources varying from 30 to 60, showing superiority of LTRED.

Fig. 6 Queue size variation

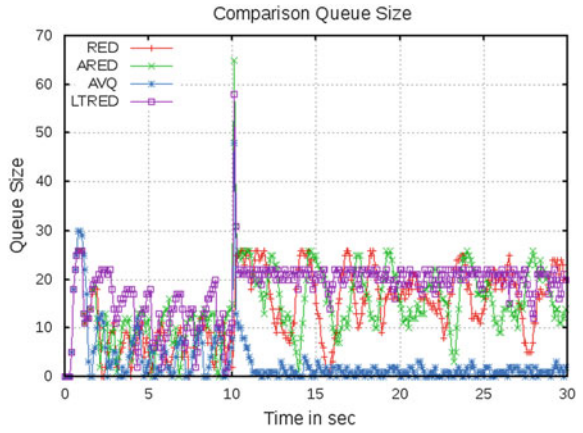


Fig. 7 Bandwidth utilization

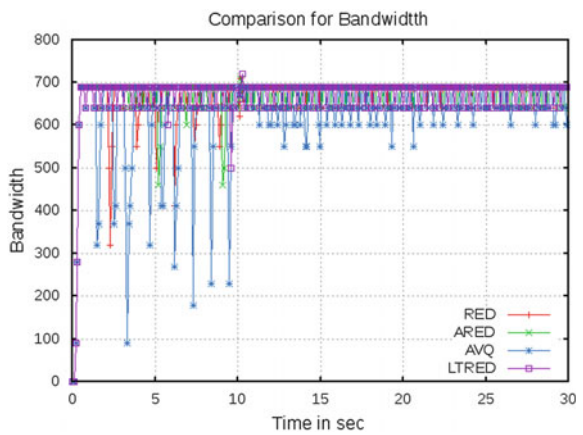


Fig. 8 Average queue size variation

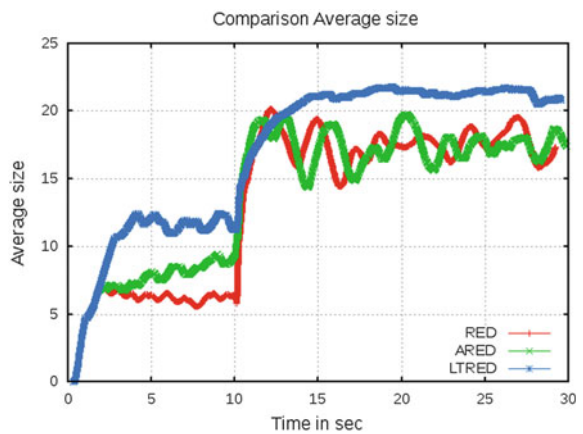


Fig. 9 RED queue sizes

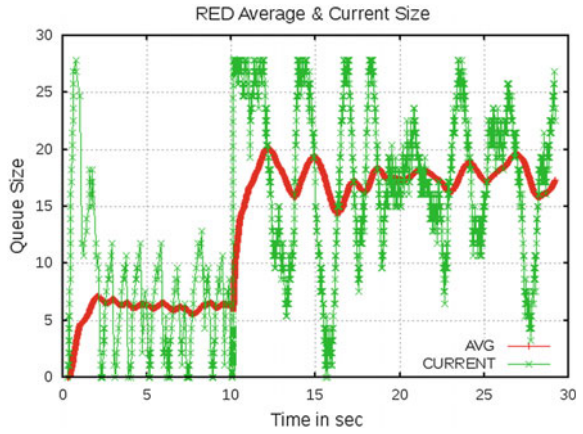


Fig. 10 ARED queue sizes

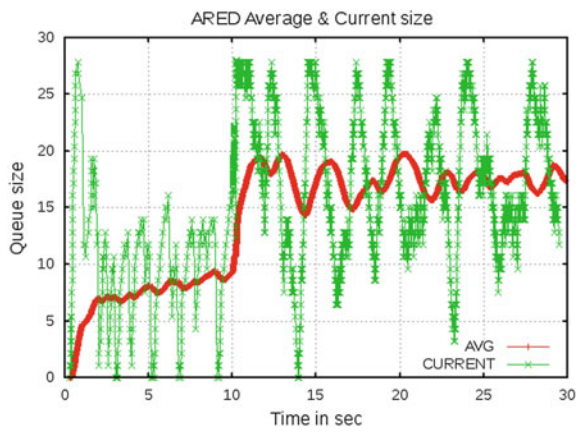


Fig. 11 LTRED queue sizes

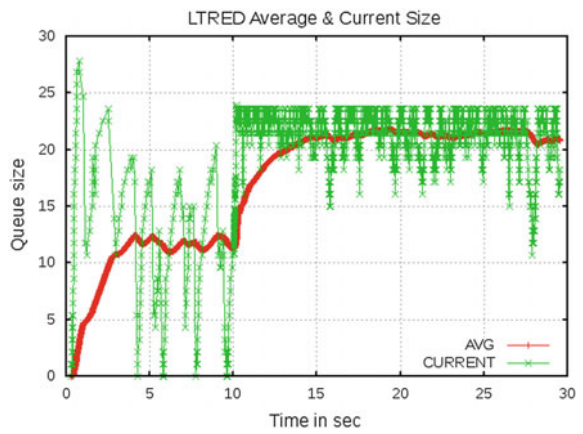


Table 1 Various simulation results for 30 sources

AQM scheme	Packets lost (%)	Packets delivered (%)	Goodput (%)	Retransmitted packets (%)	Still to be retransmitted (%)
RED	15.56	84.19	76.79	47.52	52.47
ARED	15.56	84.14	73.75	66.74	33.25
AVQ	17.73	82.23	71.31	61.58	38.42
LTRED	13.43	86.15	76.09	72.11	27.88

Table 2 Various simulation results for 40 sources

AQM scheme	Packets lost (%)	Packets delivered (%)	Goodput (%)	Retransmitted packets (%)	Still to be retransmitted (%)
RED	18.34	81.36	73.52	42.74	57.25
ARED	17.42	82.13	73.97	46.84	53.15
AVQ	20.40	79.49	70.91	42.05	57.94
LTRED	15.60	83.98	75.26	55.92	44.08

Table 3 Various simulation results for 50 sources

AQM scheme	Packets lost (%)	Packets delivered (%)	Goodput (%)	Retransmitted packets (%)	Still to be retransmitted (%)
RED	19.74	80.15	73.60	33.20	66.79
ARED	19.76	80.00	71.85	41.24	58.75
AVQ	22.01	77.93	67.75	46.23	53.77
LTRED	17.40	82.24	71.78	60.10	39.89

Table 4 Various simulation results for 60 sources

AQM scheme	Packets lost (%)	Packets delivered (%)	Goodput (%)	Retransmitted packets (%)	Still to be retransmitted (%)
RED	22.51	77.23	68.62	38.25	61.75
ARED	21.37	78.41	69.70	40.69	59.30
AVQ	23.34	76.62	67.48	39.15	60.84
LTRED	19.03	80.60	70.34	53.92	46.08

4 Conclusion

In this work, we have proposed an AQM, LTRED which is sensitive to input traffic and also tuned threshold parameter in terms of buffer size, comparing it with RED, ARED and AVQ have demonstrated better performance in terms of less number of packet loss, effective congestion indication, high throughput, reduction in mismatch behavior of the average size and the current size and high goodput values due to which retransmission of packets also reduces a lot, and effective buffer utilization at

router. An adaption mechanism based on input traffic is designed to drop the packet. The key concept is that as traffic load changes and queue length deviates from a threshold, dropping of packet occur.

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Kiran Chhabra received her B.E. in Computer Science from Nagpur University in 1995 and M.E. in Computer Technology and Applications from Chhattisgarh Swami Vivekanand Technical University, Bilhail in 2009. She is pursuing her Ph.D. in Computer Science and Engineering in congestion avoidance area. Currently, she is working as Associate Professor in Computer Science Department at MM College of Technology, Raipur.

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