

A Fair Access Mechanism Based on TXOP in IEEE 802.11e Wireless Networks

Marjan Yazdani¹, Maryam Kamali², Neda Moghim^{*3} and Mahdieh Ghazvini⁴

^{1,2,3}Department of Information Technology Engineering, Faculty of Computer Engineering, University of Isfahan, Isfahan, Iran

⁴Computer Engineering Department, Faculty of Engineering, Shahid Bahonar University of Kerman, Kerman, Iran ,
marjan.yazdani@live.com, m.kamali36@yahoo.com, n.moghim@eng.ui.ac.ir, mghazvini@uk.ac.ir

Abstract: IEEE 802.11e is an extension of IEEE 802.11 that provides Quality of Service (QoS) for the applications with different service requirements. This standard makes use of several parameters such as contention window; inter frame space time and transmission opportunity to create service differentiation in the network. Transmission opportunity (TXOP), which is the focus point of this paper, is the time interval during which a station is allowed to transmit packets without any further contention. As the fixed amounts of TXOPs are allocated to different stations, unfairness appears in the network. And when users with different data rates exist, IEEE 802.11e WLANs face the lack of fairness in the network. Because the higher data rate stations transfer more data compared to the lower rate ones. Several mechanisms have been proposed to solve this problem by generating new TXOPs adaptive to the network's traffic condition. In this paper, some proposed mechanisms are evaluated and according to their evaluated strengths and weaknesses, a new mechanism is proposed for TXOP determination in IEEE 802.11e wireless networks. The new algorithm considers data rate, channel error rate and data packet lengths to calculate adaptive TXOPs for the stations. The simulation results show that the proposed algorithm leads to better fairness. It also achieves higher throughput and lower delays in the network.

Keywords: IEEE 802.11e, MAC, TXOP, Fairness

1. Introduction

IEEE 802.11 is a set of standards for the implementation and communication of computers in wireless local area network (WLAN) in the 2.4, 3.6 and 5 GHz frequency bands.

802.11e committee is responsible to provide Quality of Service (QoS) in wireless networks and a mechanism called HCF¹ is proposed for this purpose. HCF has two access methods[1]:

- Enhanced Distributed Channel Access (EDCA² or EDCF³)
- Hybrid Coordination Channel Access (HCCA)

An important feature of HCF is the existence of four access category (AC) queues and eight traffic stream (TS) queues in the MAC layer. When a frame arrives at its MAC layer, it is tagged with a traffic priority identifier (TID), considering its QoS requirement. TIDs can take the values of 0 to 15 and the frames with TID values of 0 to 7 are mapped in to four ACs. Frames placed in these queues use EDCA access rules. On the other hand, frames with TID values of 8 to 15 are mapped into eight TS queues and use HCF controlled channel access rules. The reason of separating TS queues from AC queues is to support strict parameterized

QoS at TS queues while prioritized QoS is supported at AC queues[1].

Another important feature of HCF is the transmission opportunity (TXOP). TXOP is the time interval during which a particular STA is permitted to transmit packets without contention. Frames that are transmitted by a station in each TXOP are separated by SIFS⁴. The TXOP is called either EDCF-TXOP, if it is obtained by winning a successful EDCA contention or polled-TXOP, if it is obtained by receiving a QoS CF-poll frame from the QoS-enhanced AP (QAP). The maximum value of TXOP is called *TXOP Limit*, which is determined by QAP [1].

2. EDCA

EDCA is designed to support the contention-based prioritized QoS. Figure 1 shows the structure of EDCA.

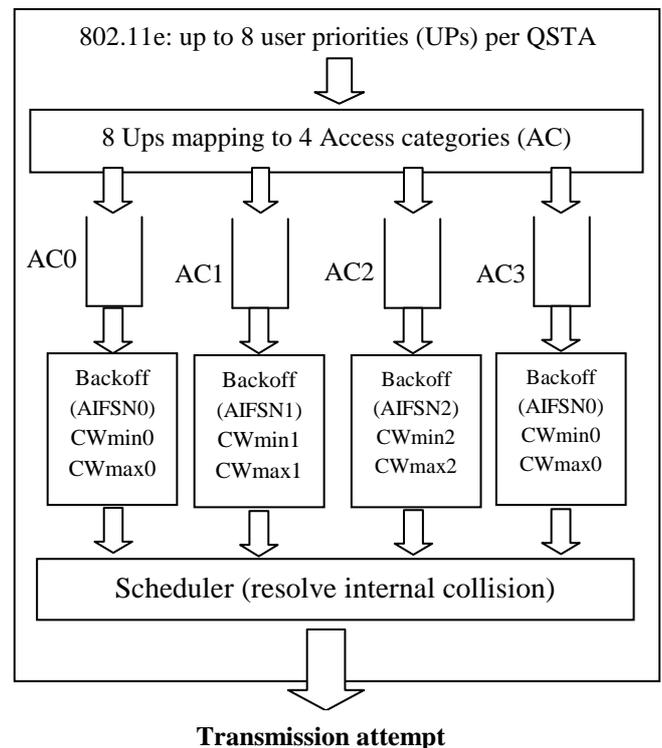


Figure 1. EDCA proposed by 802.11e [1]

Each QoS-enhanced STA (QSTA) has 4 access category queues (ACs) to support 8 user priorities (UPs). Therefore, one or more UPs are mapped to the same AC queue [1, 2].

* Corresponding author. Tel.: +98 313 7935607

¹Hybrid Coordination Function

²Enhanced Distributed Channel Access

³Enhanced Distributed Channel Function

⁴Short Inter Frame Space

Table 1 shows the mapping between ACs and UPs. These eight kinds of applications do not usually transmit frames simultaneously. Therefore, this mapping is very useful and MAC layer overhead is reduced. Fewer queues are also necessary for the implementation of ACs compared to those that are used for the UPs implementation. Each AC queue works as an independent DCF⁵ STA and uses its own back off parameters [1, 2].

Table 1.User Priority (UP) and Access Category (AC) Mapping [1]

UP	802.1D Designation	802.11e AC	Service type
2	Not defined	0	Best Effort
1	Background (BK)	0	Best Effort
0	Best Effort (BE)	0	Best Effort
3	Excellent Effort (EE)	1	Video Probe
4	Controlled Load (CL)	2	Video
5	VI (Video <100ms latency and jitter)	2	Video
6	VO (Video <10ms latency and jitter)	3	Video
7	Network Control (NC)	3	Video

Two main methods have been introduced in EDCF to provide different levels of Quality of Service[1]:

- Using different Inter Frame Space (IFS) sizes for different ACs.

- Allocation of different CW^6 sizes to different ACs.

Simulations have shown that although internal collisions are reduced in EDCF, external collisions between different QSTAs with the same priorities are still high.

CW can alter between a minimum and maximum value and it is doubled after each unsuccessful transmission attempt until a pre-defined maximum value of CW_{max} is reached. After each successful transmission, CW is reset to a fixed minimum value of CW_{min} [3].

The default values of $AIFSN[AC]$, $CW_{min}[AC]$, $CW_{max}[AC]$ and $TXOP\ Limit[AC]$ are announced by the QAP⁷ in beacon frames, and IEEE 802.11e standard allows the QAP to adapt these parameters dynamically according to the network conditions.

To improve throughput, EDCF packet bursting can be used. It means that once a QSTA gains an EDCF-TXOP, it is allowed to send more than one frame without contending for the medium any more. After accessing the medium, QSTA can transmit several frames till the channel access time does not exceed the TXOP limit bound. SIFS is used between packet bursts so that no other QSTA interrupts the packet bursting and if collision occurs, the EDCF bursting is ended. This mechanism can increase throughput by multiple transmissions, using SIFS⁸ and burst acknowledgements. It can also reduce the network overhead [1, 2].

⁵ Distributed Coordination Function

⁶ Contention Window

⁷QoS Access Point

⁸ Short Inter Frame Space

2.1 EDCA parameters

There are some EDCA parameters that can be adjusted to create different levels of service in IEEE 802.11e wireless networks. These parameters are[4]:

- Contention Window
- The arbitrary inter frame space(AIFS)
- TXOP

Table 2 shows the default EDCA parameters for different ACs.

To make more service differentiation, different CW sizes are considered for different ACs. The difference of CW_{max} and CW_{min} should not be too high for the two higher priorities ACs (voice and video) because this will lead to increased delay in the network. Due to the delay sensitivity, dropping packets is preferred to waiting for a transmission opportunity, when the network is congested [4].

Table 2. Default EDCA parameters for each AC [4]

AC	AIFS N	CW_{min}	CW_{max}	TXOP limit 802.11 a PHY (ms)	TXOP limit 802.11 b PHY (ms)
Priority 0 AC_V0	2	$\frac{aCW_{min} + 1}{4} - 1$	$\frac{aCW_{min} + 1}{2} - 1$	1.504	3.264
Priority 1 AC_V1	2	$\frac{aCW_{min} + 1}{2} - 1$	aCW_{min}	3.008	3.016
Priority 2 AC_BE	3	aCW_{min}	aCW_{max}	0	0
Priority 3 AC_BK	7	aCW_{min}	aCW_{max}	0	0

When CW is small, delay will become less and more transmission opportunities will be provided. However, a small CW causes a higher collision probability. When the number of high priority traffic streams increases, the effect of CW becomes smaller, because more collisions occur between the high priority flows [4].

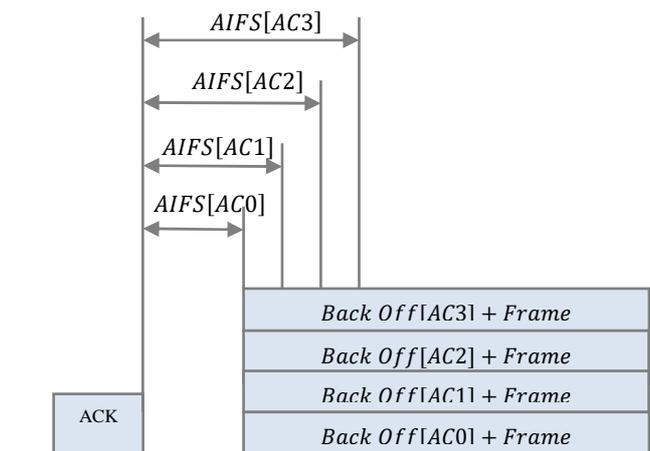


Figure 2. Channel access mechanism of the 802.11e EDCA scheme[5]

AIFS is a new inter-frame space time that is different for each AC. It is the minimum time that the medium remains idle before starting a back off. The arbitrary inter-frame space number (AIFSN) is used to calculate AIFS. AIFSN shows the number of slots that a station should wait after SIFS and before back off or before starting its transmission.

Back off time for each AC is the sum of AIFS and a random number between zero and CW. First of all, CW is set equal to CW_{min} for each AC and after each collision, CW is doubled until it reaches CW_{max} [6].

Increasing AIFS decreases the system throughput because stations must wait longer to access the medium. This effect becomes stronger when network load increases, because AIFS occurs after each transmission. Thus, larger AIFS has negative effects on the network under heavy load condition. So, it should be kept as small as possible.

TXOP is a bounded time interval which is given to each station. During this interval, each station can send as many frames as possible without any competition with other stations.

There are two kinds of TXOP:

- 1- The TXOP limit used in HCCA, which is called HCCA TXOP limit. HCCA TXOP is unique for each QSTA and it is based on the requirements of QSTA.
- 2- The TXOP limit used in EDCA, which is called EDCA TXOP limit. EDCA TXOP limit is announced in the beacon frames that are sent periodically by the access point. It has a deterministic value for each access category and is different from the TXOP of the other access categories.

The focus of the present study is on EDCA TXOP limit.

When there are multi-rate transmissions in IEEE802.11e WLANs, fixed TXOP leads to unfairness. If a fixed TXOP is considered for all stations of an AC, the stations with higher data rates can transfer more data compared to the ones with lower data rates. The reason is that the number of transmitted packets in any given period of time depends on the data rate. To solve this problem, several mechanisms have been proposed to generate TXOPs adaptive to network traffic conditions. In this paper, the performance of some of these mechanisms will be evaluated and compared. Also, a new mechanism is proposed that helps to improve the drawbacks of the evaluated approaches. In section III, a number of adjusting TXOP algorithms are introduced and their advantage and disadvantages are discussed. With the disadvantages in mind, a new method is proposed in section IV to improve the network performance. The simulation of the proposed method and its results are presented in Section V and section VI is devoted to the conclusion of paper.

3. Related works

In the adjusting TXOP algorithms, different solutions are proposed to assign larger TXOPs to the stations with lower data rates compared to the high data rate ones and in this way, they want to provide fairness in the network.

It is shown that equalizing the channel access time will lead to the throughput adaptation with the nodes' transmission rates in a multi-rate WLANs [7].

The authors in [8] introduced a Dynamic TXOP (DTXOP) algorithm which enhances fairness between upstream and downstream resource allocations in Wi-Fi networks.

The authors in [9] proposed another dynamic algorithm with the same name of DTXOP that TXOP is periodically updated for each AC according to the traffic conditions. Their simulations showed that DTXOP maintained fairness between upstream and downstream flows and improved throughput and delay as well.

In [10], TXOP is periodically adjusted according to the present traffic condition of each AC by calculating the number of stations involved in each AC and packet loss rates for each connection.

Min et al. [11] proposed a dynamic TXOP that is adjusted according to the condition of the stations' queues.

In TBD-TXOP [11, 12] method, TXOP will be equal to its default value as long as the queue length is less than the threshold. But if the queue length exceeds the threshold, TXOP will be increased. The value of new TXOP should not be too large because large TXOPs often cause large fluctuations in the performance and unfairness will occur.

Feng et al.[13] set TXOPs by using a RED like mechanism. Queue length that is a reflection of network load is used for TXOP adjustment in this algorithm. RED is a method of buffer management, in which packet loss probability increases linearly with the average queue length. Traffic conditions are monitored in QAP and stations. Similar to RED mechanism, if queue length is less than its low threshold, the lower value is assigned to TXOP and if the queue length is more than this threshold, TXOP increases linearly with the average queue length. If the queue length is greater than the upper threshold, the maximum TXOP will be used. These algorithms have focused on improving the QoS of video streams, similar to [14].

Through simulation, Suzuki et al.[15] showed that suitable TXOP determination has the ability of improving audio and video quality in the presence of channel errors.

Some simulation results and numerical analyses have shown that TXOP value should be chosen according to the buffer size [16-22].

The authors of [23] designed a distributed approach, in which each node measures its throughput in a time window. Then, it compares its throughput with the desired one and accordingly determines its TXOP value.

A dynamic multi-step TXOP allocation is presented in [24], based on the estimation of channel conditions. In this method and in each step, traffic is re-prioritized based on the network conditions and requirements of the traffic delays. Then, the new TXOP value will be adjusted with the estimation of channel error, collision and successful transmission probabilities.

The author of [25] proposed a distributed TXOP allocation scheme based on the delay bound of multimedia traffic. In the proposed scheme, a station checks the delay bound of each data packet in the queue, and allocates its TXOP value to guarantee their delay bounds.

In [26], according to the dynamisms of WLAN networks and the number of nodes in the network, a game theoretic approach called GTXOP is proposed to determine TXOP dynamically. GTXOP is defined based on the analytical models of EDCA. In GTXOP, nodes can choose their TXOP autonomously and users' QoS and overall network performance are both improved.

In [27], a method called (DTAF) is proposed for the dynamic allocation of TXOP to obtain fairness in multi-rate 802.11e networks. The proposed method estimates the network traffic conditions using the frequency of collision occurrence and TXOP will be adjusted regarding the amount of competition in the network. Simulation results showed a better fairness and also a less number of attempts to retransmit in heavy traffic loads. But this paper considered

the network in saturation condition with only three stations in the single-hop mode.

In [28], a method called adaptive opportunity (ATXOP) is proposed to solve the unfair problem in multi-rate IEEE 802.11e networks. In this algorithm, an average total data rate in the network is calculated and then the ratio of current data rate of stations to this average data rate is calculated. According to this ratio that is lower or higher than one; TXOP will be increased or decreased. Although this algorithm could solve the problem of unfairness, packet size is not considered for the TXOP determination.

Simulation results showed that this algorithm does not provide fairness for the nodes with different packet sizes. In fact; this algorithm leads to fairness only for the small packet sizes not for the large packets. Also, the network is considered error-free and the number of nodes is not considered in TXOP calculation as well. Actually, collision probability has been ignored in the network. Therefore, when the number of nodes increases, collision will increase and data traffic sent will decrease.

In [29], some parameters are defined to assign new TXOPs to the stations with different data rates. These parameters are defined considering the successful transmissions during a single TXOP in a multi-rate IEEE 802.11e WLAN. The length of the packets is considered for TXOP calculation in [29] and the simulation showed that this algorithm provides almost better fairness when different lengths of data packets are used. Similar to the [28], the network is considered error-free in [29] and the number of nodes is not considered for the TXOP calculation. So, when the number of nodes increases, collisions will more frequently occur and data traffic sent will decrease.

In another investigated algorithm [30], channel collision and error probability are predicted and considered in TXOP calculation. Thereby, the throughput enhances and the delay decreases in the network. But different data rates are not involved in the calculation of new TXOP. Therefore, fairness is not provided in a multi-rate condition. Nodes' TXOPs are increased for almost the same amount that can only improve the network throughput.

4. Proposed algorithm

In the proposed algorithm, a new TXOP determination formula is given to improve fairness and throughput together in different network conditions. This formula takes almost all the effective factors in to consideration for the TXOP calculation. The following parameters are defined in this regard:

To consider the impact of different transmission rates on TXOP determination, RF_j and $f_{ATXOP}^i [i]$ are defined [28].

$$RF_j = \frac{data_rate_j}{Avg(data_rate)} \quad (1)$$

Equation (1) represents the ratio of station j's current data rate to the average possible data rates in the network.

$$f_{ATXOP}^i [i] = \frac{TXOP [i]}{RF_j} \quad (2)$$

According to equation (2), a station with a lower data rate has a better chance of channel access compared to the stations with higher data rates. If data rate is the only assumed parameter for TXOP determination and packet size

is disregarded, nodes with different data rates will achieve unequal access to the medium. In fact; this algorithm leads to fairness only for the small packets not for the large ones. In addition, channel condition information should also be considered for better network performance.

To apply the effect of channel condition on the TXOP determination P_s is used. P_s can be calculated according to Equation (3), with the use of P_c and P_e . P_c stands for the error probability due to the collision, and P_e shows the probability of channel error [5].

$$P_s = \frac{\#SuccessfulTransmits}{\#AttemptedTransmits} = (1 - P_c)(1 - P_e) \quad (3)$$

To calculate the probability of channel error, the network is run several times and the bit error rate is determined for each node. The average of these rates is considered as the error probability of each node.

Collision probability is estimated as the ratio of the number of busy slots due to the others' transmission to the total number of slots. As it is obvious, collision probability depends on the number of nodes and it is estimated as shown in paper [19].

The proposed dynamic TXOP allocation is done in each station according to the channel condition. When a new flow arrives at a station, it starts with the estimation of successful transmission probability (P_s). Then, the estimated probability P_s will be compared to the threshold of δ . This threshold is assumed to consider the channel status and is determined experimentally. Using a comprehensive simulation, the optimum value of 0.54 is obtained for δ . If P_s probability is larger than the threshold, TXOP will be increased and if P_s is smaller than δ , TXOP will be decreased. The pseudo-code of dynamic TXOP allocation procedure is given below:

$$\begin{aligned} & \text{if } (P_s > \delta) \\ & TXOP_i = (1 + P_s) \times f_{ATXOP}^i [i] \\ & \text{else if } (P_s \leq \delta) \\ & TXOP_i = (P_s) \times f_{ATXOP}^i [i] \\ & \text{New TXOP} = \\ & \quad \left[(1 - \alpha) \times Past\ TXOP_i + \alpha \times TXOP_i \right] \times \left(\frac{PacketSize}{1000} \right) \end{aligned}$$

Each time a new TXOP is calculated according to P_s and the previous amount of TXOP. The smoothing factor (α) is equal to 0.83 and a proportion of packet length is considered in the equation to consider the impact of different packet sizes on the TXOP determination. Regarding Equation (2), different data rates are considered and according to the pseudo-code, channel error rate and data packet lengths are also involved in the new TXOP calculation.

Therefore, all of the previously discussed effective factors are considered in the proposed algorithm and it is expected to achieve better throughput and delay using the proposed algorithm, compared to the other methods.

5. Simulation

OPNET is used to simulate and evaluate the performance of the proposed algorithm. Its performance will be compared with that of the standard IEEE 802.11e EDCA. Three fixed nodes with different data rates of 2Mbps, 5.5Mbps and 11 Mbps are simulated that send data to the same fixed destination.

TXOP of the stations with AC1 and AC0 traffic is assumed equal to MSDU⁹ and each time they send a single data unit disregarding their data rate. Therefore, (AC2) video and (AC3) voice Traffics are considered in this simulation. OPNET parameters are set according to Table 3. It is noteworthy that each simulation was run for 10 different Seeds. The average of the results is shown in the following figures.

Table 3.Simulation parameters

Network Size	300m×300m
Start Time (s)	Constant(0)
ON State Time (s)	Constant(60)
OFF State Time (s)	Constant(10)
Inter arrival Time (s)	Exponential(0.0041)
Traffic Type of Service	Interactive Voice Or Interactive Multimedia
Physical Characteristic	Direct Sequence
Short Retry Limit	7
Long retry Limit	7
Transmit Power	0.005
Data Rate	1,2,5.5,11 Mbps
AC3(CW _{min} ,CW _{MAX})	(7,15)
AC2(CW _{min} ,CW _{MAX})	(15,31)
AC1(CW _{min} ,CW _{MAX})	(31,1023)
AC0(CW _{min} ,CW _{MAX})	(31,1023)

Using the proposed algorithm, the TXOPs calculated for AC2 and AC3 in different data rates and packet sizes are shown in Table 4.

Table 4: New TXOPs for AC3 and AC2

Packet size (Bytes)	New TXOP for AC3 (voice) in ms			New TXOP for AC2 (video) in ms		
	2048	1024	300	2048	1024	300
2 Mbps	52.46	26.23	7.68	65.53	48.34	14.2
5.5 Mbps	19.8	9.9	2.9	36.49	18.25	5.34
11 Mbps	10.47	5.23	1.53	19.29	9.65	2.83

The AC2's data traffic sent, achieved by our proposed algorithm, is shown in Figures 3 and 4 in different data rates. Data packets are considered 1024 and 300 bits long in Figure 3 and 4 respectively.

The AC3's data traffic sent, achieved by our proposed algorithm, is shown in Figures 5 and 6 in different data rates. Data packets are considered 1024 and 300 bits long in Figure 5 and 6 respectively. These improvements are because of considering data rate, channel error rate and data packet lengths in the TXOP calculations.

Figures 3-6 show that traffics with different data rates in each access category have almost the same chance to send

data traffic. Therefore the proposed method achieves better fairness. As mentioned before, packet length is considered for the TXOP calculation in the proposed algorithm. Therefore, as it was predictable, Figures 10-13 confirmed that this algorithm is relatively fair for different packet lengths.

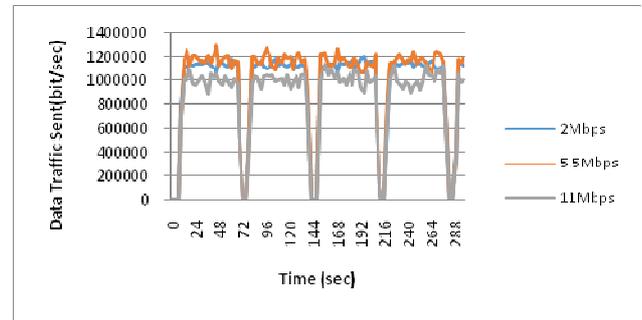


Figure 3. Data Traffic Sent at AC2 for the data packets of 1024 bits long

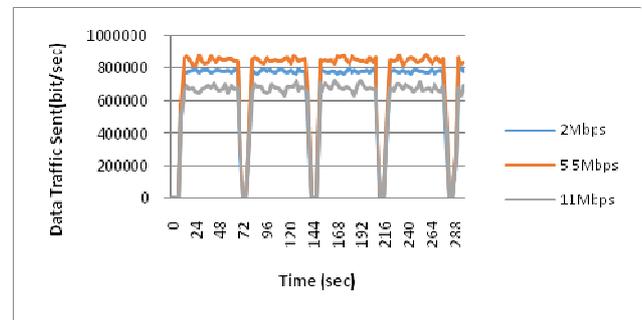


Figure 4.Data Traffic Sent at AC2 for the data packets of 300 bits long

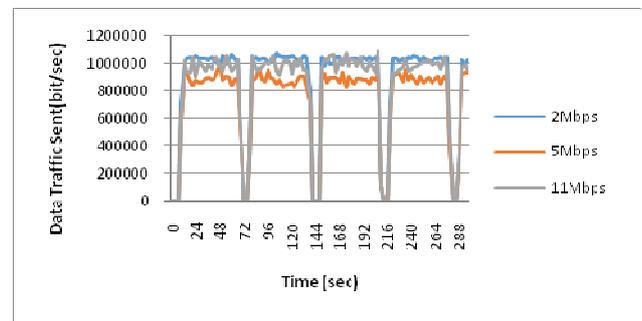


Figure 5. Data Traffic Sent at AC3 for the data packets of 1024 bits long

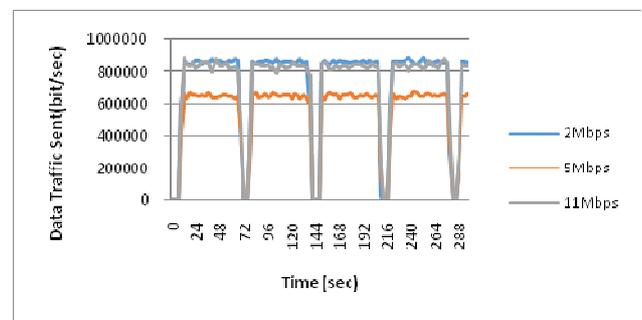


Figure 6. Data Traffic Sent at AC3 for the data packets of 300 bits long

The average throughput is equal to the total number of bits (in bits/Sec) forwarded from wireless LAN layers to the higher layers in all the WLAN nodes. Figures 7 and 8

⁹ MAC Protocol Data Unit

suggest that the throughput of the network in AC2 and AC3 classes of traffic has been increased through using the proposed algorithm in comparison with fixed TXOP usage in EDCA standard. This means that the proposed algorithm has improved both the throughput and fairness in the network.

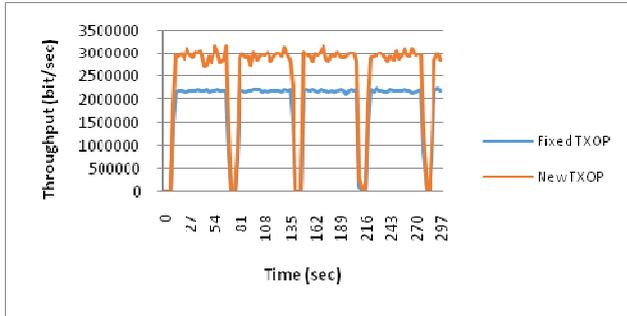


Figure 7. Throughput comparison of the fixed TXOP and the proposed TXOP at AC2

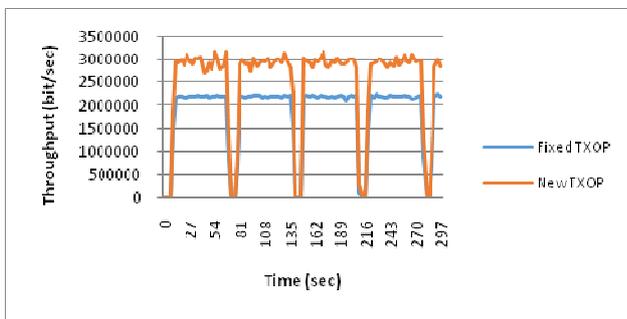


Figure 8. Throughput comparison of the fixed TXOP and the proposed TXOP at AC3

The comparison of these methods in terms of the end-to-end delay is shown in Figures 9 and 10. The end-to-end delay refers to the time taken for a packet to be transmitted from the source to the destination. It is noteworthy that the end-to-end delay does not include the delays of the lost packets, which are dropped due to the successive collisions. However, end-to-end delay resulted in the proposed algorithm is smaller than the delays resulted from EDCA at both AC2 and AC3 traffic classes.

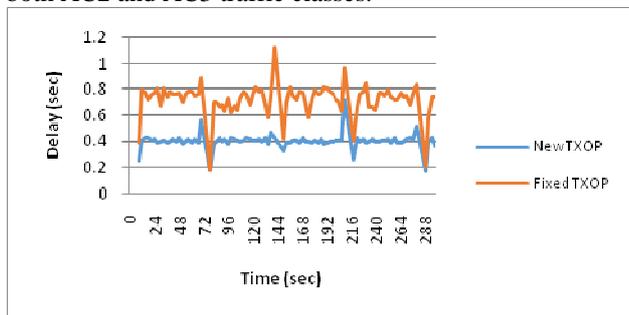


Figure 9. Delay comparison of the fixed TXOP and the proposed TXOP at AC2

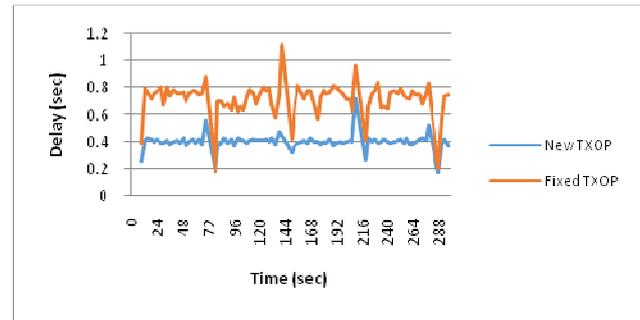


Figure 10. Delay comparison of the fixed TXOP and the proposed TXOP at AC3

6. Conclusions

When different data rates are used in IEEE 802.11e WLANs, unfairness problem occurs. The reason is that equal TXOP or transmission opportunity is allocated to the stations, without any data rate consideration. The stations with higher data rates can send more traffic in the network compared to the lower rate ones.

Several algorithms have been proposed to solve this problem. They determine the new TXOP adaptive to the network's traffic condition. Some of these protocols were investigated in this paper and their advantages and disadvantages were discussed.

Considering the drawbacks, an algorithm was proposed that takes different effective factors in to account for the TXOP determination. The simulation results showed that the proposed algorithm leads to better fairness and network throughput. Furthermore, packets' delay improves.

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