Performance comparison of Downlink Packet Scheduling Algorithms in LTE Network

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Abstract—Long Term Evolution (LTE) was introduced by the Third-Generation Partnership Project (3GPP) and is considered as the latest step towards the fourth generation of radio technology. This paper investigates the performance of well-known packet scheduling algorithms such as Proportion Fair (PF), Maximum Largest Weighted Delay First (M-LWDF), Exponential Proportion Fair (EXP/PF), Frame Level Scheduler (FLS), Exponential rule (EXP rule), and Logarithmic rule (LOG rule) in terms of delay, throughput, and packet loss ratio (PLR) by using the LTE-Sim open source simulator. Different traffic types are used, and Simulation results show that in video traffic, FLS and EXP algorithms provide a higher system throughput compared to other algorithms while keeping the delay and packet loss ratio small. However, in the case of best-effort traffic, results show a high delay and PLR with low throughput. The main contribution of this paper is to determine the appropriate downlink scheduling algorithm for VOIP, video, and best-effort traffics in 3GPP LTE.

Index Terms--- downlink scheduling; LTE; OFDMA; QoS; video traffic

I. INTRODUCTION

Due to the increasing number of wireless cell phone users and the Internet creating an increased traffic volume, as well as bandwidth scarcity [1], the existing networks became unable to satisfy their users, forcing telecommunication companies and researchers to find a way or to develop solutions that can improve the performance of cellular communication networks. One of these solutions is LTE which was introduced by the 3GPP and is considered to be the latest step towards the fourth generation of radio technology.

LTE supports carrier bandwidths starts from 1.4 MHz and goes up to 20 MHz, and increases to 100 MHz in LTE Advanced. LTE introduces a high throughput with low latency and at a low cost. While Universal Mobile Telecommunication Systems (UMTS) and High Speed Packet Access (HSPA) still use circuit-switching for voice calls and the IP network for other data services such as internet access; in contrast, LTE uses a simple architecture with all IP networks [2].

The aim of this paper is to investigate the performance of the well-known packet scheduling algorithms in the downlink 3GPP LTE system, such as PF, EXP/PF, M-LWDF, FLS, LOG rule, and EXP.
for uplink, while two sub-frames are left as a special sub-frame. Therefore, TDD can work in

<table>
<thead>
<tr>
<th>Config. number</th>
<th>Sub-frame number</th>
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<tbody>
<tr>
<td>0</td>
<td>D S U U U D S U U</td>
</tr>
<tr>
<td>1</td>
<td>D S U U D S U D D</td>
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<tr>
<td>2</td>
<td>D S U U U U D D D</td>
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<td>3</td>
<td>D S U U U D D D D</td>
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<td>4</td>
<td>D S U U D D D D D</td>
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<td>5</td>
<td>D S U D D D D D D</td>
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<tr>
<td>6</td>
<td>D S U U U U D S U U</td>
</tr>
</tbody>
</table>

D = downlink sub-frame; U = uplink sub-frame; S = special sub-frame.

In OFDMA, the full frequency bandwidth is divided into orthogonal subcarriers, subcarriers where each subcarrier is allocated 15 kHz. The LTE frame consists of 12 consecutive subcarriers and 10ms duration. Each frame consists of 10 sub frames; each sub-frame is 1ms, which is equal to the Transmission Time Interval (TTI); and then each sub-frame is equal to two time slots, where each slot is 0.5 ms in the time domain and 12 subcarriers in the frequency domain. However, each slot is composed of a resource block (RB), which is the minimal radio resource allocation unit in the LTE [5], [6] (see Fig. 2).

Each RB consists of seven symbols when the normal Cycle Prefix (CP) is used or six symbols when the extended CP is used, such as the evolution Multimedia Broadcast Multicast Service (eMBMS) sub-frame.

III. DOWNLINK PACKET SCHEDULING ALGORITHMS

The evolved NodeB (eNodeB) is responsible for sharing available RBs between users depending on some rules and respecting the resource allocation strategies, which play a fundamental rule in increasing spectrum efficiency and system performance [7], since maximizing the system throughput is one of the most important challenges in the design of the downlink 3GPP LTE system packet scheduling algorithms. At each TTI, and for each sub-carrier carrier, each user sends feedback about their downlink channel’s channel condition to its eNodeB [8] (see Fig. 3).

In fact, each time and frequency domain has an effect on the channel’s quality for many reasons, for example, the effects of fading, multipath propagation, the Doppler effect, and so on. So, different users experience distinctive Channel Quality Indicator (CQI) at different times. For this reason, in OFDMA, channel aware solutions are used to give a higher priority to those users who are experiencing a better channel condition.

One of the RRM main functions is packet scheduling, that is, the smart assignment of a user to use the available system RBs while taking into the account the many QoS parameters needed to satisfy the performance matrices. The designing of packet scheduling algorithms in the LTE network has become more difficult and complex with the need to support the variable QoS requirements of different traffic, while raising the efficiency of the system as much as possible.

However, there are many techniques carried out in the LTE network that attempt to satisfy the end users and network providers by performing a high cell capacity, decreasing packet delay and packet loss while maintaining fairness between users.
following subsection, we will briefly restate and discuss the most well-known downlink packet scheduling algorithms in the LTE network.

A. Proportional Fair Scheduling

One of the most recognized packet scheduling algorithms is Proportional Fair (PF), which assigns free system resources to a user whose average feedback CQI is high. It allocates user $j_{mn}$ in RB $m_n$ in any given sub-frame $f$, if:

$$j_{mn} = \arg \max_{j=1,...,J} \frac{R_j(m,f)}{T_j(f)}$$  \hspace{1cm} (1)

where $R_j(m,f)$ is the achievable rate by user $j$ in RB $m$ and the sub-frame $f$ [9].

B. Modified Largest Weighted Delay First

Another well-known scheduling algorithm is the modified largest weighted delay first (MLWDF) which was developed to support a range of data users with a variety of QoS requirements. A user is selected according to the following equation [10][14]:

$$U_i = \arg \max_{j=1,...,J} \alpha_i \frac{W_j(t)}{u_j}$$  \hspace{1cm} (3)

$$\alpha_i = -\log b_i \frac{1}{r_i}$$  \hspace{1cm} (4)

where $W_j(t)$ denotes the Head of Line (HOL) packet delay, $\alpha_i$ denotes the weight factor, $r_i$ is the delay threshold for user $i$, and $\delta_i$ is the acceptable packet loss rate for user $i$.

C. Frame Low Scheduling

Frame Level Scheduling (FLS) focuses on the QoS for a video multimedia application in the downlink side. Two levels scheduling have been used to design an FLS algorithm, both upper level and lower level. In the upper level, discrete time linear control theory is exploited. In the lower level, a proportion fair scheduling algorithm (PF) is used. In other words, FLS works together with PF, where FLS works in the upper level with frames to decide how much data should be transmitted by each resource block; while in the lower level, PF has been used to maintain fairness and keep system throughput at the maximum as possible. However, according to the FLS rules, the best-effort flows can be served just after all the video flows have been served [11].

D. Exponential Rule and EXP/PF

The adaptive Exponential PF rule (EXP/PF) was first developed for multimedia traffic in the TDD system over the OFDMA system. It uses channel state and queue information explicitly and could offer a streaming service as well as best-effort data services to mobile users [12]. EXP rule was implemented to support video streaming services to guarantee specific delay constraints. In EXP/PF, the properties of PF and the exponential function of the end-to-end delay are both taken into account. EXP/PF can work with video and best-effort applications. The metric for video can be calculated as:

$$m_{i,k}^{EXP/PF} = \exp \left( \frac{\alpha_i D_{HOL,i} - x}{1 + \sqrt{x}} \right) \frac{d_i(t)}{R^i(t - 1)}$$  \hspace{1cm} (5)

where $x = \frac{1}{N_T} \sum_{j=1}^{N_T} \alpha_i D_{HOL,j}$ and $N_T$ denotes the number of active downlink video applications, $\alpha_i$ as in eq. 4, $R^i(t - 1)$ is the last average throughput achieved by the user $i$ until time $t$ and $d_i(t)$ is the expected data-rate for the user $i$ at time $t$ on the RB $k$ [13].

E. The Logarithmic Rule

The logarithmic rule (LOG rule) is one of most well-known scheduling algorithms in the LTE system as it proposed a radical sum-rate monotone (RSM), firstly was proposed in [14]. The LOG rule algorithm has some policies that deal with both the mean delay and robustness. For the LOG rule algorithm, the following equation is used to calculate the metric:

$$m_{i,k}^{LOG\text{-rule}} = b_i \log(c + \alpha_i D_{HOL,i}) \cdot I_{i,k}^t$$  \hspace{1cm} (5)

where $b_i$, $c$, and $\alpha_i$ are tunable parameters; and $I_{i,k}^t$ is the spectral efficiency for a user $i$ on the sub-channel $k$. 

IV. SIMULATION

A. Simulation Setup

We have evaluated the performance of different scheduling algorithms using the LTE-Sim simulator, which is an open source simulator developed by Telematics Lab at the Electrical & Electronics Engineering Department, the Technical University of Bari [15]. Simulation parameters are listed in Table II.

The simulation was ran 150 times to evaluate well-known algorithms such as PF, EXP/PF, MLWDF, FLS, EXP rule, and LOG rule, in terms of delay, packet loss ratio (PLR), and throughput, using different types of traffic. For each algorithm, a varying number of users and different traffic was experimented.

The experiment starts with 4 users and is performed 5 times repeatedly, and then the average was taken in order to ensure accurate results. After that, the same scenario was conducted with 8, 12, 16, and 20 users. Each user has 1 video flow with a 242 bit rate, 1 VOIP flow, and 1 best-effort flow. Moreover, a high number of users has been used to
evaluate the performance of the well-known algorithms in term of packet loss.

In fact, it takes a long time to run 150 experiments with different scenarios (30 scenarios, each scenario was repeated 5 times) and calculate the average delay, PLR, and throughput. However, by using the powerful tool of Shell Script, it becomes easier to perform many experiments with different scenarios in one click.

<table>
<thead>
<tr>
<th>TABLE II. SIMULATION PARAMETERS</th>
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<tbody>
<tr>
<td>Carrier frequency             2GHz</td>
</tr>
<tr>
<td>Downlink bandwidth            5MHz , 25 RBs</td>
</tr>
<tr>
<td>Symbols for TTI               14</td>
</tr>
<tr>
<td>Sub-frame Length              1ms</td>
</tr>
<tr>
<td>Number of eNodeB              1 eNodeB</td>
</tr>
<tr>
<td>eNodeB radius                 1 km</td>
</tr>
<tr>
<td>eNodeB power transmission     43 dBm</td>
</tr>
<tr>
<td>Modulation Scheme             QPSK, 16QAM, 64QAM</td>
</tr>
<tr>
<td>Number of UEs                 4, 8, 12, 16, 20; 10-100</td>
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<tr>
<td>UE speed                      3 km/h</td>
</tr>
<tr>
<td>Application flow              1 VOIP, 1 Video, 1 best-effort</td>
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<tr>
<td>Video rate                    242 kbps</td>
</tr>
<tr>
<td>Simulation time               46 ms</td>
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</table>

B. Simulation Results

1. Delay:

As shown in Fig. 4 and Fig. 5, in case of VOIP traffic, a high delay was found when the FLS algorithm was used while the other algorithms provided low delay. In contrast, the FLS achieved the lowest delay compared to other algorithms when video traffic was used.

2. Throughput

As shown in Fig. 6, all algorithms resulted in almost the same QoS requirement for VOIP services. In Fig. 7, in the case of video traffic, it can be observed that the FLS resulted in the highest throughput, followed by the EXP rule algorithm; while the LOG rule, MLWDF, and the EXP/PF displays almost the same QoS requirement as video streaming services. In PF, the throughput decreased as the number of users increased and exceeded 8 users.

As illustrated in Fig. 8, in the case of best-effort traffic, the system throughput decreased when the FLS algorithm was used. This phenomena occurred because in the FLS algorithm, RBs are reserved to the video stream first, then the free RBs are reserved to other traffic.

3. Packet Loss

As shown in Fig. 9 and Fig. 10, in the case of video traffic, it can be observed that both the FLS and
the EXP rule algorithms achieved low packet loss ratio while the other algorithms provided a higher PLR. Actually, the PLR increase in all algorithms as the number of users increase.

V. CONCLUSION

In this paper, the performance of the most well-known packet scheduling algorithms in the LTE system were evaluated in the downlink side. Their performances are evaluated using an open source simulator called LTE-Sim. Simulation results demonstrated that, in the case of using video streams, the FLS and EXP rule algorithms achieved high throughput with low delay and PLR; and achieved low throughput with high delay and PLR in the case of using non video applications.

As for future work, we are thinking of working on mechanisms that support high throughput with low latency to run smoothly the Evolved Multimedia Broadcast Multicast Service (E-MBMS), through reduction of the feedback rate and efficient selection of the optimal Modulation and Coding Scheme (MCS) level based on its standard deviation.

REFERENCES