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Network Planning Problems in 3G/4G Wireless Systems

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Abstract

Wireless network planning is a very complex process, the result of which influences on the success of network operators. A poorly planned network cannot achieve the required Quality of Service. It also involves extra costs and fewer benefits for its network operator. Actually, wireless network planning deals with a large number of different aspects. Since there has been evolution in wireless communications towards the third generation (3G) and forth generation (4G) wireless systems some Core Network planning problems for the 3G wireless networks and some Radio Access Network planning problems for the 4G wireless networks are discussed in this paper. The problem of performance estimation of UMTS Release 5 Internet Protocol Multimedia Core Network System (IM-Subsystem) nodes is considered in details.

Keywords

UMTS, 3G Core Network, 4G Radio Access Network, IP-Subsystem, self-similarity

Working Group N 3

1. Introduction

There has been an evolution in wireless communications almost every ten years. The first generation (1G) in 1980s and the second generation (2G) mobile systems in 1990s have been oriented mainly for providing circuit-switched (CS) services to users. The 2G subscribers have used the rate for data transfer up to 14 kbp/s as a maximum. In 1996, European Telecommunications Standards Institute (ETSI) decided to enhance 2G GSM standard in annual Phase 2+ releases that incorporate the third generation (3G) features such as General Packet Radio Service (GPRS) and Enhanced Data Rates for GSM Evolution (EDGE). The data rates for users of the systems are limited to less than several hundreds of kbp/s.

Universal Mobile Telecommunications System (UMTS) as the 3G mobile system will be introduced during first decade of new century. It is specified by ETSI and the world-wide 3G Partnership Project (3GPP) within the framework defined by the International Telecommunication Union (ITU) and known as International Mobile Telecommunications-2000 (IMT-2000). The 3G systems can support 2 Mbp/s for indoor environments and at least 144 kb/s for vehicular environments.

ETSI and 3GPP are introducing UMTS in phases and annual releases. UMTS Rel'3 (sometimes called as Rel'99) is a 3G GSM successor standard using the GSM Phase 2+ enhanced core network (CN). The most important evolutionary step toward UMTS is to introduce a packet switched core network (PS CN) domain. The main function of the PS CN domain is to support all services (GPRS, WAP, etc.) provided to both GSM subscribers and UMTS users [1].

The following phases after Rel'3 specify how voice and multimedia can be supported by IP technology. It is characterized by creating of the Internet Protocol (IP) Multimedia Core Network Subsystem (IM-subsystem), which comprises all PS CN domain elements for providing telecommunication services within UMTS Rel'4,5,6. The IM-subsystem contains a uniform way to maintain Voice over IP (VoIP) calls and offers a platform to multimedia services. The examples of IM services are voice telephony, real-time interactive games, videotelephony, instant messaging, emergency calls, multimedia conferencing [2]. In the UMTS Rel'5,6 all traffic coming from Radio Access Network (RAN) to the CN is supposed to be all IP based [1].

The next step of wireless communications evolution is the fourth generation (4G) of mobile communication systems (the systems beyond IMT-2000). Now it is difficult to predict when the 3G evolution ends and the 4G really starts [1]. The 4G systems should offer significantly higher bit rate than 2 Mb/s, have high capacity with a low bit cost and be able to support all type of telecommunication services from the viewpoint of multimedia communications [3]. It is supposed that on the CN side of the 4G systems the main purpose is to minimize changes and utilize the 3G CN elements and the 3G CN functionality as much as possible [1]. The CN development is summarized in the Table I.

TABLE I. CORE NETWORK DEVELOPMENT

WIRELESS SYSTEMS	CORE NETWORK DOMAINS
2G	CS CN
2G phase 2 +	CS CN and PS CN
3G (UMTS Rel'3)	CS CN and PS CN (enhanced 2G phase 2+ CN)
3G (UMTS Rel'4)	CS CN, PS CN, IM CN
3G (UMTS Rel'5,6)	IM CN
4G	IM CN (enhanced 3G CN)

There are some important features of the global evolution process in wireless communications.

The 3G wireless systems should be designed to support for a high-speed transfer of a large amount of multimedia information between users. One of the main properties of the data traffic in the 3G systems is a large diversity depending on the profile of services provided to 3G users. It is expected that the traffic in the 3G systems will expand considerable [4]. The growing data/multimedia traffic leads to increasing the total load on network subsystem elements. Moreover, traffic patterns generated by 3G services may be quite different from traditional Poisson models used for circuit switched voice traffic. When modeling packet-switched multiservice networks it is necessary to take into account the notion of self-similarity [5-8]. Due to the high variability of burstiness of the traffic, the use of the classical teletraffic theory for a performance evaluation of PS CN domain elements may give essential faults; in particular, the network parameters can be underestimated. Such faults are unacceptable when IM-subsystem planning as well, therefore, principles of the teletraffic theory cannot be applied in this case.

Due to above reasons, the following 3G network planning problems occur:

- the prediction problem of a demand for 3G services;
- the estimation problem of 3G data traffic parameters;
- the problem of the performance evaluation of IM-subsystem nodes taking into account the self-similar nature of the multiservice traffic.

One more feature of the evolution in wireless communications concerns changes in the 4G wireless systems. It is seen from the Table I that the CN evolution is quite temperate. From the viewpoint of functional capabilities the enhanced CN of the 3G systems will be able to support 4G services [1]. So, it is expected that the RAN will undergo the main changes, in particular, on a physical transmission layer. In the 4G systems, because there will be a need to deal with the enormous amount of traffic, the Base Station (BS) radius cell is supposed to be shorter than that of 3G systems [9]. Therefore, the 4G RAN will comprise more BSs and more frequent handover will occur resulting in a heavy load on the links between such elements of RAN as Remote Network Controller (RNC) and BSs, suggesting changes to the RAN architecture [3,9,10]. It may raise valuable funds for the RAN deployment.

For the 4G RAN a new and innovative topology of physical links configuration has been proposed and analytically argued with the viewpoint of a load and routing capabilities only

[3,9,10]. The topology is called the “cluster-cellular” or, in other words, it is known as a ring one. In such topology BSs are grouped in a “cluster” and there is a “cluster-main” BS connected to the RNC. BSs in a cluster may be connected to each other by optical fiber links that are preferred as the dominant links to construct the 4G RAN, first of all, from the viewpoint of link capacity [11]. It should be emphasized that the infrastructure cost of optical fiber networks is very expensive because of high installation cost. For this reason, it is very important to analyze the ring topology of the RAN physical links with respect to its reliability and cost, to compare it with other topologies and give recommendations on applying of different ones in the 4G RAN. Besides, it is worthwhile to determine the optimal configuration of physical links between BSs in the ring topology to decrease the 4G RAN deployment cost.

Taking into account the considerations above, the research should be concentrated on two subjects. The first part of the research is focused mainly on the 3G CN planning problems. The second part of the research deals with the 4G RAN topological planning problems.

The main objective of the research is to develop solution methods for the above-mentioned problems. It will enable planning 3G/4G networks in such a way that both technical and economical advantages can be achieved when constructing and exploiting the networks.

In this paper one of the main problems of Core Network planning is considered in details. This is the problem of performance evaluation of IM-subsystem elements. This problem arises because of the fact that the traffic generated by 3G services may be self-similar or long-range dependent in nature (i.e., bursty over a wide range of time scales).

Self-similarity is observed in different networks; in particular, in local area networks [12], Internet [13], wireless networks [5] and others. It is shown in [7] that in GPRS in the case of aggregated traffic and also in the case of individual WAP and WEB traffic traces, the results strongly suggest long-range dependency (values of the Hurst parameter are about 0.8). Besides, the packet arrival process of WAP and WEB traffic may be considered as a class of processes consisting of the superposition of an infinite number of ON/OFF-sources. Through the characterization of the sum of the covariances, it is possible to establish a simple explicit necessary and sufficient condition for the process to be long-range dependent [14]. It is reasonable to suppose that self-similarity may occur in 3G wireless networks as well. This is in sharp contrast to commonly made traffic modeling assumptions, because self-similarity is characterized by stronger dependence of a variance from time than linear dependence [7]. The traffic does not smooth out in the case of aggregation, leading to congestion situations and packet-drops due to the burstiness of the traffic. In the case of self-similar traffic the applied methods for performance analysis and network dimensioning are different from those applied to statistically more simple traffic, which can be modeled with Markovian processes [15,16]. For example, the queue tail behavior is heavy-tailed in the case of self-similar input traffic [17].

Thus, the use of the classic teletraffic theory for a performance evaluation of packet multiservice network elements gives essential faults, in particular, network parameters may be underestimated [13,18,19]. In literature [17,20,21] the approaches of overcoming such sort of difficulties are considered. In our research the results from [20] concerning self-similar multiservice traffic is developed and applied for evaluating probabilistic and time characteristics of such IM-subsystem element as Gateway GPRS Support Node (GGSN) for the UMTS Rel'5 IM-subsystem. The method for GGSN performance evaluating is based on using the FBM/D/1/W queueing system.

2. UMTS Rel' 5 core network architecture aspects

The reference architecture for UMTS Rel' 4 and Rel' 5 from 3GPP TR 23.821 is the same [1]. In the development of UMTS Rel' 5 the focus has shifted to the PC CN domain, which has been extended with IM-subsystem functionality. The vision of UMTS Rel'5 from the All IP point of view taken from [1] is shown in Fig. 1. As seen from Fig. 1, the principle of allocation of data flows between end users and GGSN leads to increasing of the load on the network elements while approaching to GGSN. The GGSN is the node that the most exposed to the self-similarity influence in UMTS. The most important events determining the load on GGSN on the network level are arriving IP packets. Currently, a transport technology for delivery of IP packets to/from GGSN is not defined uniquely. For instance, ATM may be applied as one of the possible cases of such technology [19].

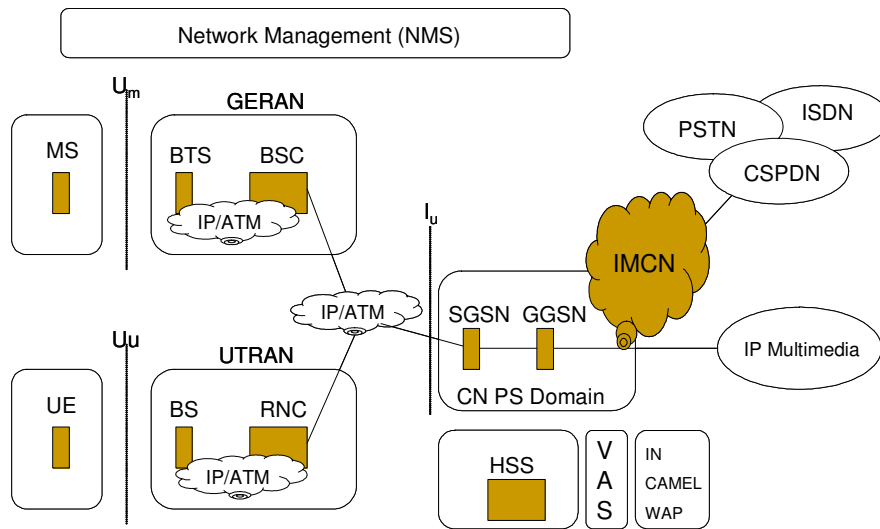


Fig. 1. Vision of UMTS Rel' 5 (all IP)

3. The load model

Let us assume that values $s(t)$ of a random process with interdependent increments are total load arrived to the node (server) in the time point $t > 0$. Current values $s(t)$ in the time interval $[0, t)$ may be determined by the information units number. If the corresponding process is ordinary then an increment is one information unit. The increments intensity is the rate parameter λ , 1/sec. Realizations of process $s(t)$ are non-decreasing step functions with increments taking place in random time points.

Let us consider a random variable $S_T = s(t=T)$, $T \in [0, t)$ that is a sample of a random process $s(t)$. By definition S_T is a sum of interdependent identically distributed random variables. If $E(S_T) = \lambda T \gg 0$ then conditions of the central limiting theorem are fulfilled. Here, $E(\cdot)$ is operator of statistical averaging. Accordingly, S_T may be approximated by a Gaussian random variable [23]. Taking into account abovementioned assumptions S_T may be defined as:

$$S_T = \lambda T + \sqrt{b(T)} \cdot x, \quad (1)$$

where $x = N(0,1)$ is a normalized Gaussian random variable with the zero mean and the unit variance, $b(T)$ is a variance of S_T .

If $b(t) = \sigma^2 t$ and $t > 0$ then the univariate probabilities distribution of the process $s(t)$ coincides with the corresponding distribution of a Brownian motion process or a displaced Wiener process [24,25]. Similarly, the process $s(t)$ corresponds to a Poisson process if condition (1) is fulfilled when x is a Poisson random variable and $b(t) = \lambda t$.

There are different ways of self-similarity load modeling [12,20,21]. With reference to (1) self-similarity may be taken into account as

$$b(T) = (\sigma^2 T)^{2H}, \quad 0.5 \leq H < 1, \quad (2)$$

where H is the Hurst parameter. Expressions (1) and (2) specify a model of a total traffic load arriving to a server input by a time point $t=T$.

4. The queueing system model

We assume that $s(t)$ arrives to the server input. The server is modeled by queueing system with deterministic rate C , 1/sec and the buffer size $(W-1)$, $1 \leq W < \infty$. The queueing system is the stable one because there is a stationary probability distribution if $C > \lambda$. In accordance with the Kendall's notation for queues the system is $G/D/1/W$ [28]. The corresponding system may be also defined as $FBM/D/1/W$ [20] if the expressions (1,2) are fulfilled. Here, the FBM is a normalized fractional Brownian motion, i.e. the corresponding process is a strictly self-similar one.

5. The task estimation definition

When stability conditions are fulfilled the average value of the total load arrived to the queueing input by the time point $t = T > 0$ is less than the queueing system can serve for the same time interval. It should be emphasized that the load is a random process. Therefore, it is possible to appear an event when the buffer will be overflow. The probability of the event is defined by statistical properties of an unserved traffic process that may be written as

$$V(t) = \max[0, S(t) - Ct] \quad (3)$$

The introduction of operator $\max [0, x]$ in (3) is caused by nonnegative values of an unserved load. It is similar to the introduction of an adsorbing barrier in the coordinate origin point for the displaced self-similar (fractional) Wiener process.

The estimation problem is to determine values of parameters C and W . It should be done by taking into account the following condition. The probability that the unserved load will be greater than the parameter W must not exceed the preset threshold ε :

$$P[V(t) > W] = \varepsilon, \quad t > \varepsilon \quad 0 < \varepsilon \ll 1 \quad (4)$$

6. The estimation task solution

Taking into consideration the approximation of the random process $s(t)$ sample by the random Gaussian variable defined by the expressions (1,2) we have the lower bound for the buffer saturation probability

$$P[V(t) > W] \geq \max_{T>0} P[x > \alpha(T)], \quad (5)$$

where $\alpha(T) = [(C - \lambda)T + W] / (\sigma^2 T)^H$.

The expression (5) shows that the probability of events union is not less than the probability of each event. Taking into account that the random variable X is the normalized displaced Gaussian random one, the expression (5) may be transformed as

$$P[V(t) > W] \geq \max_{T>0} \left[\int_{\alpha(T)}^{\infty} \exp(-x^2/2) dx / \sqrt{2\pi} \right] \approx \max_{T>0} [0.5 \exp(-\alpha^2(T)/2)] \quad (6)$$

Let us take into consideration the logarithmic function monotony for the expression that is equivalent (4,6). Then the expression binding the parameters C , W , λ and the buffer saturation probability ε is

$$-\ln \varepsilon \approx \min_{T>0} \frac{((C - \lambda)T + W)^2}{(\sigma^2 T)^{2H}} \quad (7)$$

The solution of the equation (7) may be found by the parameter T differentiation and equating of the obtained derivative with zero [20]. It gives the following expression

$$T_m = \frac{WH}{(1-H)(C-\lambda)} = \arg \min_{T>0} \frac{((C - \lambda)T + W)^2}{(\sigma^2 T)^{2H}} \quad (8)$$

Substituting T_m in (7) and transforming the expression we finally get

$$C/\lambda = 1 + n \left(\frac{\sqrt{-2 \ln \varepsilon} W^{H-1} H^H}{(1-H)^{H-1}} \right)^{1/H}, \quad 0.5 \leq H < 1, \quad (9)$$

where $n = \sigma^2 / \lambda$.

Substituting n , W , λ and ε values in (9) we get the upper bound (if $H=0.8$) and lower bound (if $H=0.5$) of the server service rate C . One of the main parameters of queueing system is the inverse parameter (9) $\rho = \lambda/C$ called the utilization factor. Fig. 2 illustrates dependences of the utilization factor from the magnitude $n = \sigma^2 / \lambda$ for various values of the Hurst parameter and the buffer capacity W when $\varepsilon = 10^{-5}$.

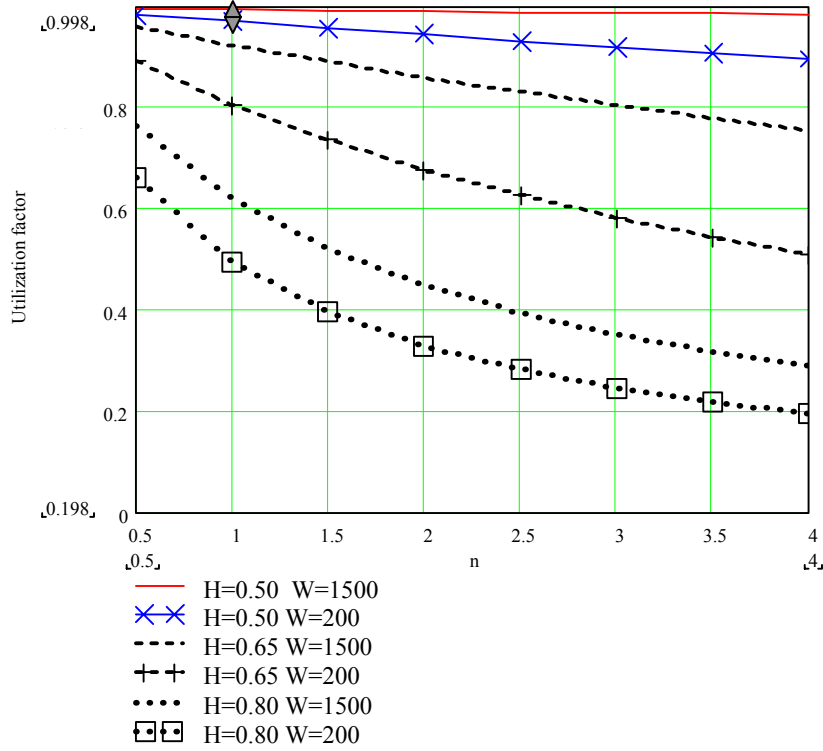


Fig.2. Self-similarity influence on the server utilization factor

Trends of the curves (Fig.2) show that it is very important to take into account the self-similarity influence while assigning server parameters. According to [26] the Hurst parameter of the corresponding Internet traffic may achieve values exceeding 0.8. It should be emphasized that the area of the dependences when $n=1$ and $H=0.5$ (as shown in Fig.2) corresponds to the case of the Poisson arrival process.

It is wise to take the value ε sufficiently small. It enables to have the acceptable probability of messages blocking arriving on the server. In this case the server buffer will be filled partly.

For determination of the upper bound of the average queue length in the server buffer the expression based on results [17,27] may be used:

$$q_{\max} = \frac{(\lambda/C)^{1/2(1-H)}}{(1-\lambda/C)^{H/(1-H)}} \quad (10)$$

The classical result for M/D/1 system may be applied for determination of the lower bound of the average queue length:

$$q_{\min} = \frac{\lambda/C}{1-\lambda/C} - \frac{(\lambda/C)^2}{2(1-\lambda/C)} \quad (11)$$

Little formula [14] enables to determine the upper and the lower bound for the average service time τ using expressions (9-11):

$$\tau_{\max} = q_{\max} / C, \quad \tau_{\min} = q_{\min} / C, \quad (12)$$

Expressions (9-12) allow estimating bounds of the probabilistic and time characteristics of the single server under the self-similarity load influence.

7. Case study

In this section the example illustrating the above-presented method is considered. Since there are no exact regulations on transport network protocols on SGSN-GGSN interface at the present moment it is assumed that ATM is used as underlying technology for delivery of IP packets. The rate of information units (ATM cells) arriving on SGSN is multiple (k) of 2 Mbit/sec. The parameters characterizing the server normal functionality may be estimated by the following way.

Let $k = 20$ and in average 30% of the channel throughput is in use during the messages delivery to SGSN. Then, the value of the intensity of ATM cells arriving on the SGSN input is $\lambda \approx 30000 \text{ s}^{-1}$. If a number of SGSNs connected to the GGSN is 4 then the total value of the intensity of ATM cells arriving to GGSN input is $\lambda \approx 120000 \text{ s}^{-1}$. In accordance with (4.8), (4.9) and (4.10) the relationships between the GGSN server capacity, the upper bound for average queue length in the GGSN buffer, the upper bound for the average service time of information units in the GGSN buffer and the parameter n ($n = \sigma^2/\lambda$) are shown in Figures 3, 4, 5 respectively ($W = 50, 200; H = 0.8; \varepsilon = 10^{-7}$).

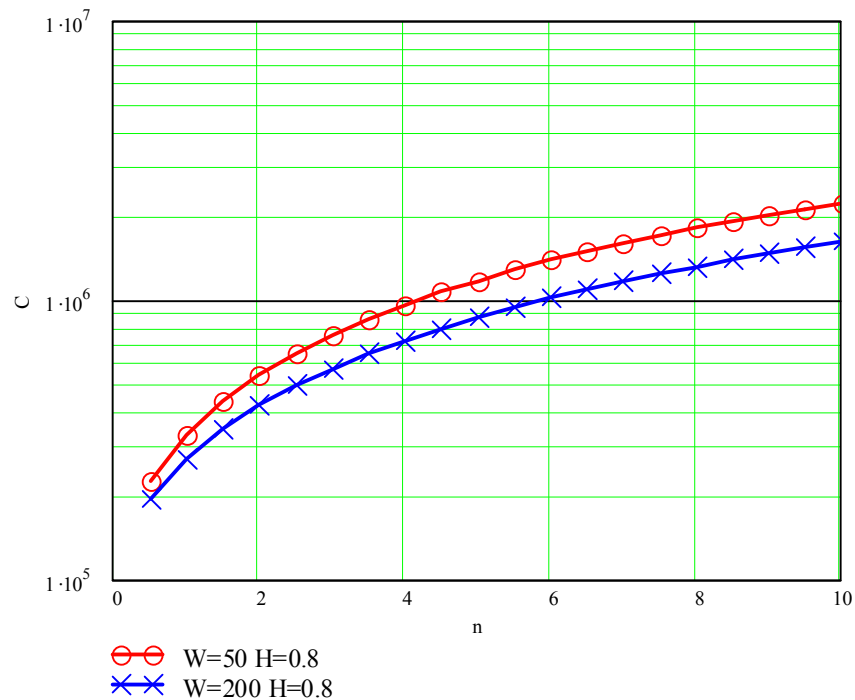


Fig.3 GGSN server capacity estimating

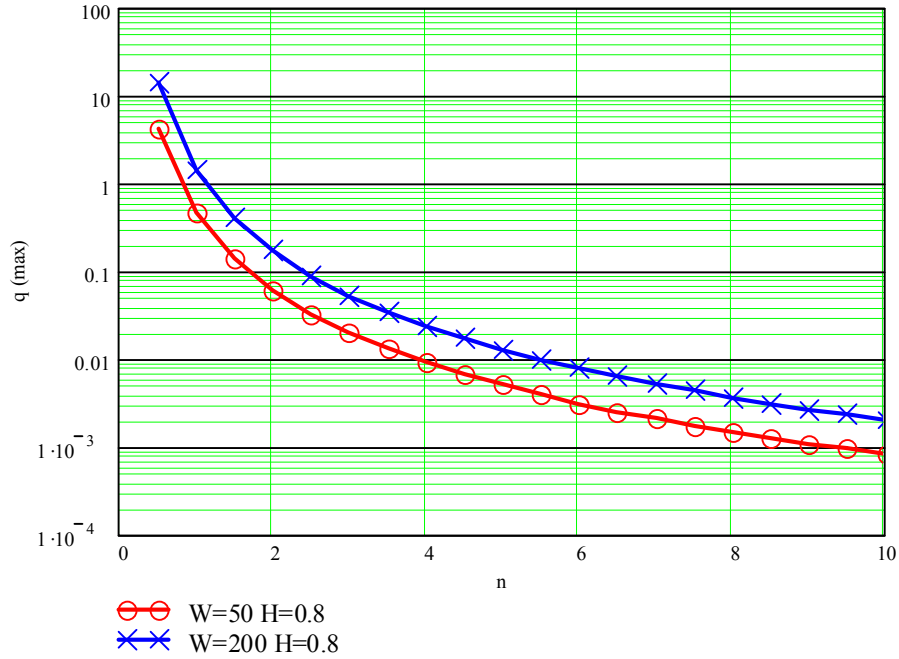


Fig. 4 The upper bound for average queue length in the GGSN buffer

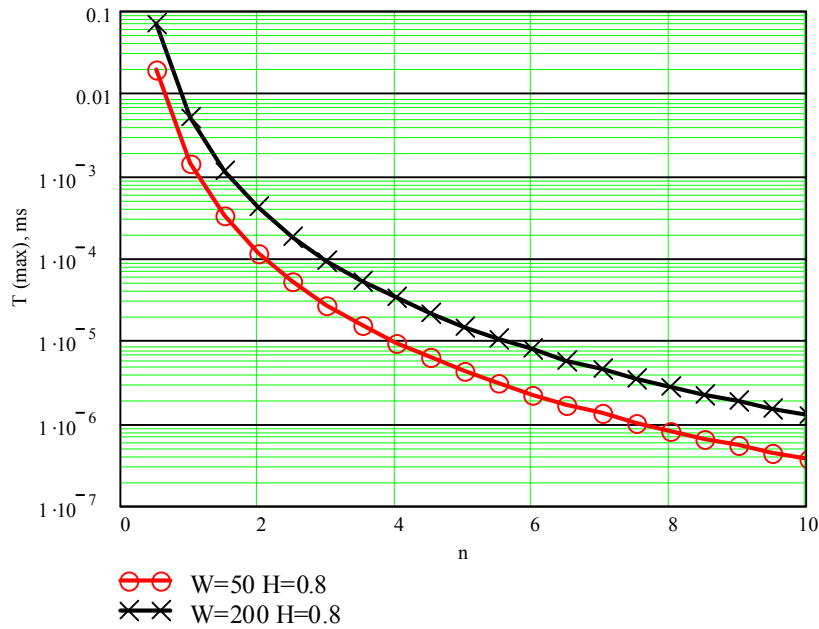


Fig. 5 The upper bound for the average service time of information units in the GGSN buffer

8. Conclusion

In this paper the influence of self-similar input on GGSN performance in UMTS Rel'5 IM-subsystem has been analyzed. FBM/D/1/W queueing system for evaluating the GGSN parameters

was applied. The submitted method enables determining the following probabilistic and time characteristics:

- upper and lower bounds for the GGSN service rate;
- upper and lower bounds for the average queue length in the GGSN buffer;
- upper and lower bounds for the average service time of information units in the GGSN buffer;
- the server utilization.

The obtained results point to a need to take into account self-similarity while assigning the GGSN parameters.

As well known, when providing multimedia services based on IP technologies one of the main aspects is to ensure Quality of Service (QoS). From this point of view the presented method may be extended for performance evaluating of other IM-subsystem elements, in particular, for SGSN performance evaluating.

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