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_____ Systèmes communicants _____



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de recherche*



Performance Analysis of an Header Compression Protocol: The ROHC Unidirectional Mode

Alain Couvreur* , Louis-Marie Le Ny[†] , Ana Minaburo* , Gerardo Rubino[†] ,
Bruno Sericola[†] , Laurent Toutain*

Systèmes communicants
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Abstract: The performance of IPv6 in the radio link can be improved using header compression algorithms. The 3GPP (3rd Generation Partnership Project) consortium in its technical specification has adopted the ROHC (RObust Header Compression) protocol of the IETF (Internet Engineering Task Force) standard track for real-time applications using RTP/UDP/IPv6 and UDP/IPv6. This paper presents the analysis of the proposed standard ROHC deployed in an UMTS radio link and discusses different schemes to increase compression performance. The results are based on our IPv6 implementation of the ROHC header compression algorithm and on a simple and accurate analytical model used to evaluate the packet loss probability.

Key-words: ROHC, Performance evaluation, IPv6, UMTS

(Résumé : tsvp)

* ENST Bretagne, BP 78, 2 rue de la Châtaigneraie, 35512 Cesson Sévigné Cedex, France.
{Alain.Couvreur}{Anacarolina.Minaburo}{Laurent.Toutain}@enst-bretagne.fr

[†] IRISA, Campus de Beaulieu, 35042 Rennes Cedex, France.
{Louis-Marie.Lenys}{Gerardo.Rubino}{Bruno.Sericola}@irisa.fr

Analyse de performance d'un protocole de compression d'entêtes: le mode unidirectionnel ROHC

Résumé : Les performances d'IPv6 sur un lien radio peuvent être améliorées en utilisant des algorithmes de compression d'entêtes. Le consortium 3GPP (3rd Generation Partnership Project) a adopté dans sa spécification technique le protocole ROHC (RObust Header Compression), standard de l'IETF (Internet Engineering Task Force), pour les applications temps-réel utilisant RTP/UDP/IPv6 et UDP/IPv6. Cet article analyse le protocole ROHC déployé sur un lien radio UMTS et étudie différentes techniques permettant d'augmenter les performances de la compression. Les résultats sont basés sur notre implémentation IPv6 du protocole de compression d'entêtes ROHC et sur un modèle analytique simple et précis utilisé pour évaluer la probabilité de perte des paquets.

Mots-clé : ROHC, Evaluation de performances, IPv6, UMTS

1 Introduction

The UMTS (Universal Mobile Telecommunication System) is the third generation (3G) mobile network under development in the 3GPP Consortium. The scenario will be of a convergence between mobile telephony and Internet, where a wide variety of services independent of the user location will be provided. UMTS is expected to provide many services based on the Internet or IP-based services. However, the use of IP will mean large overheads, and will take a significant amount of bandwidth, which is already scarce in cellular links. Moreover, the use of IPv6 flows for data transmissions in Releases 4 and 5 of UMTS will degrade the situation, as the overhead bytes in IPv6 are larger. Hence, there is a need to compress the large headers and save some valuable radio resources. In the UMTS reference protocol architecture, a special layer (PDCP: Packet Data Convergence Protocol) dedicated to header compression has been introduced. The header compression is a scheme that removes (ideally all) the redundant information from the header. Existing header schemes do not perform well over cellular links due to the high BER (Bit Error Rate), around 10^{-2} to 10^{-3} , and the high RTT (round trip time), around 200ms, of the radio channel. In this scenario, a valid solution is represented by ROHC (Robust Header Compression); it has been proposed by the IETF (Internet Engineering Task Force) ROHC working group. ROHC aims at providing a high compression efficiency and a high robustness and can be used in the PDCP layer. In order to investigate the behavior of header compression for real-time applications in UMTS architecture, we have implemented ROHC header compression and have studied it on a simulated model of the error conditions and the long round trip time that prevail in the radio link. We present ROHC header compression algorithm and show how it works with UMTS. In the following sections, we discuss the existing header compression schemes and then explain the ROHC header compression protocol. Next, we discuss the implementation of the protocol and develop an analytical model which is validated by the implementation and which allows us to compute the packet loss probability for various values of the input parameters.

2 Header Compression

The problem of the performance of the IP protocol over a low bandwidth link has been studied since 1984 with the Thin-wire protocol specified by Faber, Delp [5]. In 1990, Van Jacobson proposed a mechanism based on the header redundancy information, which compressed the 40 bytes of the TCP/IPv4 header to 3-6 bytes [8]. There are other propositions for the different protocol headers based on redundancy, as the CTCP (IP Header Compression) header mechanism [4], which manages a wide variety of streams and gives particular attention to IPv6. The CRTP (Compressing IP/UDP/RTP Headers for Low-Speed Serial Links) header mechanism [3] is a detailed specification for RTP (the basic protocol for real-time data streams) compression. Previous studies have demonstrated that the performance of these protocols is still quite poor in wireless links. The next generation of header compression mechanisms aims to add a robust and efficient compression scheme that can be used in a low bandwidth link with significant error rate. An Algorithm to improve the

performance and robustness for TCP/IP flows was proposed by Degemark [4] who suggested a strategy to recover from the loss of compression synchronization, which occurs due to high BER. A second algorithm to further reduce the probability of synchronization loss was presented by Perkins (PEHRC) [14]. These schemes were further improved by Giovanardi who proposed two retransmission strategies (ABP-SCS) and (ABP-DCS) [6], in order to reduce the TCP/IP header synchronization loss in an UMTS radio link. The new contributions for TCP/IP header compression are TAROC [12], EPIC [15] and ROHC+ [1]. TAROC's goal is the limitation of error propagation using TCP congestion window tracking. EPIC uses a variant of Huffman encoding to produce a set of compressed header formats. ROHC+ proposes a new profile and an algorithm for compressing TCP stream within the ROHC framework. Many contributions have been proposed for the IP/UDP/RTP protocol stack, like the adapted header compression for real time multimedia application (ACE) [11], the header compression using keyword packets [13], or the header compression based on Checksum (ROCCO) [10]. These three contributions are the basis for the header compression standard ROHC (Robust Header Compression) described in RFC 3095 [2].

3 The ROHC Protocol

ROHC header compression algorithm was conceived to reduce the header sizes of IP packets to be sent through a cellular link, which is characterized, by high BER, long RTT (Round Trip Time) and residual errors. In order to compress the Header, ROHC mechanism makes a classification of the header fields. This analysis is based on how the values in the header fields change during the transmission of a stream. These fields are separated and assigned to the static and the dynamic chain of the compressed header packets. Static refers to the information which remains more or less constant during the lifetime of the stream and dynamic refers to the information which may change but whose change pattern may be known. ROHC uses different header format packets to establish the information in the decompressor, see Figure 1. First, the static information is sent to the decompressor and after the compression level is increased by sending just the dynamic information or the compress header fields variation, which is the encoding value of the timestamp, RTP "more" flag and the sequence number. The principle of ROHC is to send the minimal information with high robustness. One key element here is CRC (Cyclic Redundancy Check) that is computed over the original header fields before compression. The decompressor then verifies the CRC after decompressing the header and checks whether it has received the correct information or if the information has been corrupted due to transmission errors in the link.

3.1 ROHC Context

ROHC uses a context maintained between compressor and decompressor to store the information about the header stream. This context contains the last correct update of the original header and the redundant information in the stream. This context is kept both in the compressor and in the decompressor in order to ensure the robustness of the mecha-

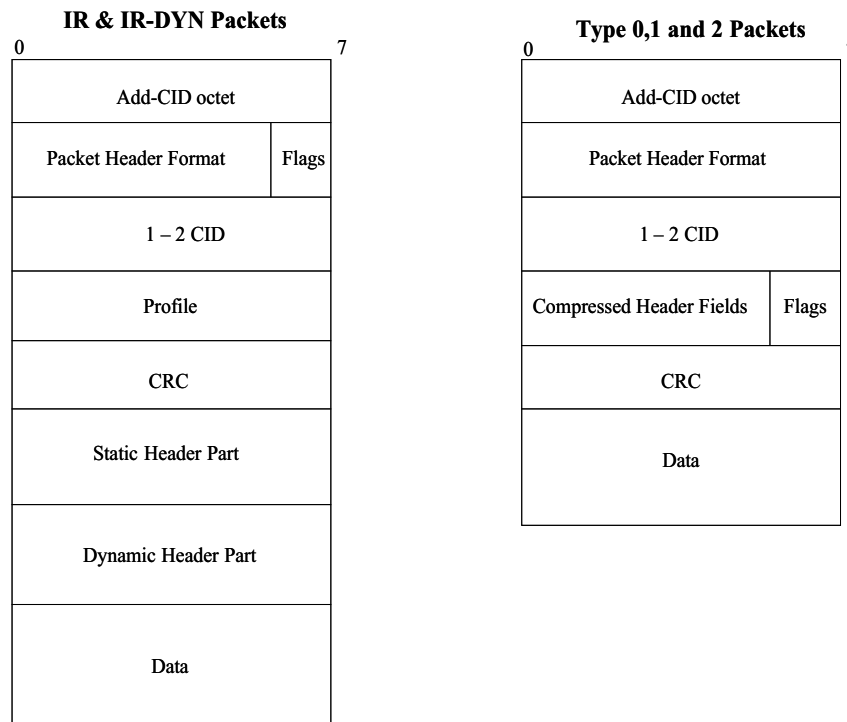


Figure 1: ROHC general header format packets. IR belongs to first compression level, IR-DYN and UOR-2 to second compression level and type 1 and 0 to third compression level.

nism. Each time a value in the context changes, the context is updated. If the context is lost due to transmission errors then there is no synchronization between the compressor and the decompressor. The decompressor then can request for the context update through the possible use of acknowledgements. Each flow in a channel has its context, which is identified by a CID (Context Identifier) which is a number that differentiate the flows in a channel, and the context in compressor and decompressor.

3.2 ROHC Profiles

The profiles are used to define different types of header streams. The decompressor can define the stream type by looking at the profile. Currently five profiles have been standardized but this could change in the future: Profile 0, without compression. When this profile is used, only the ROHC Context Identifier is added to each packet to let the decompressor know that the stream is not compressed. ROHC defines 4 profiles:

- Profile 1, IPv4/v6/UDP/RTP header compression. This is the generic profile; this profile compresses three-header protocol (IPv4/v6/UDP/RTP).
- Profile 2, IPv4/v6/UDP header compression. This is a variation from profile 1, where the compression is only applied for the UDP/ IPv4/v6 headers.
- Profile 3, IPv4/v6/ESP header compression. This profile compresses the ESP/ IPv4/v6 protocol.
- Profile 4, IPv4/v6 header compression. This profile compresses only the IPv4/v6 header.

3.3 ROHC Negotiation

The first phase of ROHC protocol is a negotiation. In this phase, the compressor and the decompressor learn about the different characteristics of the link and the parameters that they will use for compression. Negotiation is made while establishing a channel. A channel is a connection between two nodes. Each channel has a compressor and a decompressor at each side, with the possibility of it being either bi-directional or unidirectional. The present solution [9] is for an Internet network where each IP interface in the IP layer can have multiple channels, each one being bi-directional or unidirectional. The present ROHC negotiation establishes the following parameters: MAX_CID: the maximum CID that can be used. MAX_Header: the largest header that can be compressed. MRRU (Maximum Received Reconstructed Unit): when segmentation is used, it helps to know the maximal size of the segment in bytes. Sub-options: there can be zero or several sub-options. Until now, only one sub-option has been specified: The Profile. It informs about the profiles that are supported.

3.4 Compression Levels

ROHC has three compression levels: Initialization and Refresh (IR), First Order (FO) and Second Order (SO). Each compression level in each operation mode uses different header types, as shown inside the octagons in Figure 2. Forward and backward transitions to the different compression levels depend on the operation mode; there are three operation modes. The compressor always sends the header format packet that fits the information needed for decompressor.

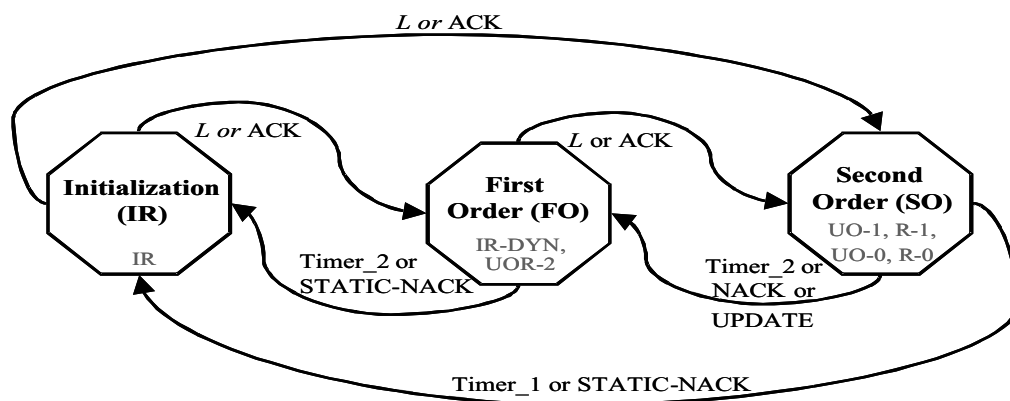


Figure 2: ROHC Compression Levels. Each operation mode has these three levels of compression. Each compression level uses different packets. For the SO level, the packets used are UO-1, R-1, UO-0, and R-0.

The size of the compressed header depends on the compression level and the header information required by the decompressor. In the first level of compression (IR), the header size is between 48 to 130 bytes; in the First Order, the compressed headers have a size between 3 to 84 bytes. In the last compression level (SO), the header is compressed up to just one byte.

3.5 Operation Modes

The different operation modes allow the compressor to change from one mode to another based on the link characteristics and the performance requirements. Each operation mode has its own behavior. In Unidirectional (U) mode the compressor operates over links where feedbacks are not possible. The compressor uses a confidence system, L to base its confidence about decompressor status. This system sends the same header format packet L times to transit forward to the next compression levels. U-mode also uses two timers and keeps on coming back to initial compression levels when timer expires or when there is an update in the header information. One timer is used to come back to IR compression level, `Timer_1 (IR_TIMEOUT)` and other is used to come back to FO compression level, `Timer_2 (FO_TIMEOUT)`. The bi-directional optimistic (O) mode is very similar to the U-mode but the decompressor can also send negative acknowledgements. This mode does not use the two timers, but it uses the confidence system, L . The compressor goes downward to the initial compression levels on the receipt of negative feedbacks. There are two kinds of negative feedbacks: NACK, that force compressor to go to FO compression level, and the STATIC-NACK that forces it to go to IR compression level, see Figure 2. If the compressor receives a header that will update the entire context or if a SO packet cannot communicate the changes then, it goes down to FO or IR level of compression. The bi-directional reliable operation (R) mode works only with the acknowledgements received from the decompressor. Each time the compressor receives an ACK/NACK; the compressor changes the compression level. It goes to the IR compression level if a STATIC-NACK is received, see Figure 2. Here, a secure reference principle is also enforced in both compression and decompression logic. The principle means that only a compressed packet carrying a seven or eight-bit CRC can update the decompression context and can be used as a reference for subsequent decompression.

3.6 ROHC Decompressor

The decompressor has a state machine based on the context state (see Figure 3). The decompressor has three states. The first state is No Context (NC), the decompressor stays initially where there is no context and reached it when the context is lost. In this state only the IR header, format packets are decompressed and any other header format packet is dropped. The decompressor changes to Full Context (FC) state as soon as correct decompression of a header takes place (verified by CRC) or if the context is established. The Static Context (SC) state is not reached except when there is an error and the dynamic part of the header is lost. The decompressor then waits for a timer to be expired in U-mode. In the other modes, it sends a NACK to the compressor for getting a FO compression level packet. The decompressor uses a "k out of n" failure rule, where k is the number of packets received with an error in the last n transmitted packets. This rule is used in the state machine of the decompressor in order to assume the damage of context and to move to the downward states after sending to compressor a negative acknowledgment, when bi-directional link is used.

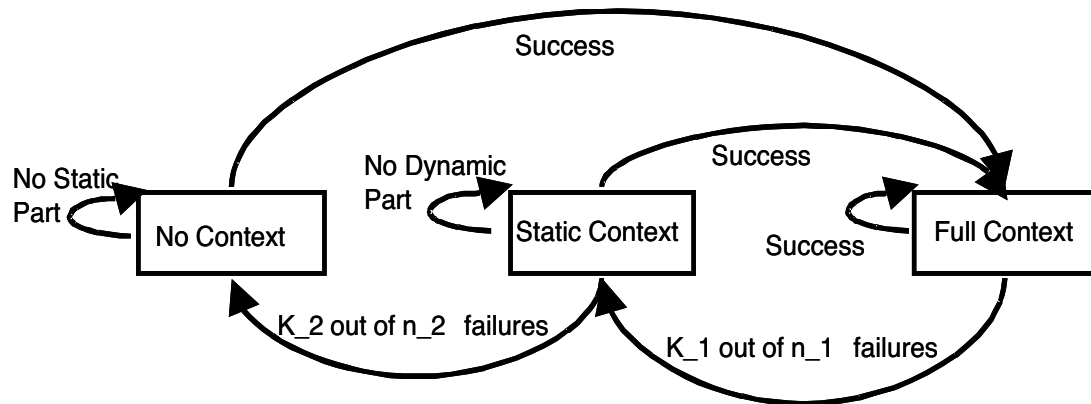


Figure 3: Decompressor State Machine.

3.7 Mode Transition

Each time the decompressor wants to work in a new operation mode, it launches the mode transition by sending an ACK/NACK with mode parameter bits set to the new operation mode. During transition, the compressor is allowed to work only in the first two compression levels, all the packets sent contain a CRC to verify the information, each of the compressor and the decompressor keeps two control variables: Transition and Mode and they deny any new request for transition till current transition gets completed. If the transition variable has a pending value, the transition mode cannot be released. To finish the transition, an ACK with the valid sequence number and the new operation mode has to be received by the compressor, if it is not the case, the transition variable keeps the pending value and Mode variable keeps the old value. In addition, to initiate the transition to the R mode, the context must have been established between the compressor and the decompressor. For the other transitions, the decompressor can start the transition mode at any moment.

3.8 ROHC compression parameters and schemes

The value of the compression parameters of ROHC that determine the efficiency and robustness are not defined in ROHC specification and are not negotiated initially but are stated as implementation dependent. The compression parameters see Figure 4 and the schemes that use them, are as follows:

1. L: In U-mode and O-mode the ROHC compressor uses a confidence variable (L) in order to ensure the correct transmission of header information.
2. Timer_1 (IR_TIMEOUT): In U-mode, the compressor uses this timer to return to the IR compression level and periodically resends static information.
3. Timer_2 (FO_TIMEOUT): The compressor also uses another timer in U-mode and this timer is used to go downward to FO compression level if the compressor is working in SO compression level.
4. Sliding Window Width (SWW): The compressor while compressing header fields like Sequence Number (SN) and Timestamp use W_LSB encoding that uses a Sliding Window of width equal to SWW.
5. W_LSB encoding is used to compress those header fields whose change pattern is known. When using this encoding, the compressor sends only the least significant bits. The decompressor uses these bits to construct the original value of the encoding fields.
6. k and n: The ROHC decompressor uses a "k out of n" failure rule, where k is the number of packets received with an error in the last n transmitted packets. This rule is used in the state machine of the decompressor to assume the damage of context and move downwards to a state after sending a negative acknowledgment to the compressor, if bi-directional link is used. The decompressor does not assume context damage

and stays in the current state until k packets arrive with error in the last n packets. The k_1 , n_1 values are used to assume dynamic context damage and k_2 , n_2 to assume static context damage.

7. MODE: There are two bits in the header format packets to define the operation mode in which the compressor and the decompressor work. There are only three-operation modes, U, O and R mode defined in [2].

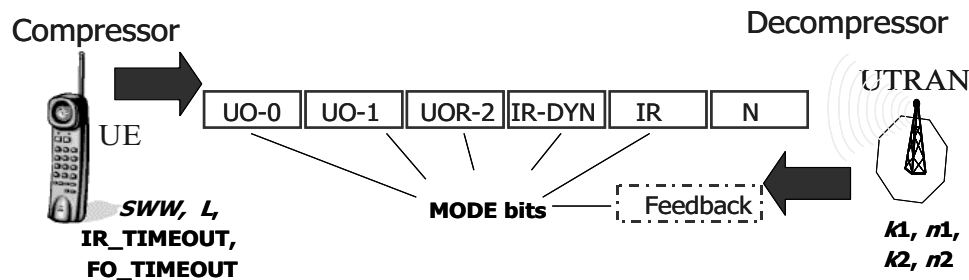


Figure 4: The Compression Parameters. They are localized in the compressor, in the decompressor and in the different header format packets. Note that at each end of the link both Compressor and Decompressor are present.

4 ROHC Compression Implementation

The ROHC implementation is the development of Profile 1 (IPv6/UDP/RTP) header compression. It returns the sequence of packets sent, the average throughput, the number of CRC failed packets in the ROHC implementation and the number of lost packets in the application, the sequence number execution, the header size in each packet sent and received and the number of each ROHC packet sent. Our experimental system consists of a video application platform in IPv6 and a PPPoEv6 based on FreeBSD4.5 with Kame. The platform is composed of a video application located in both nodes. Through the PPPoEv6, the client receives the video header compressed packets that will be decompressed by the ROHC decompressor in the other node. At the beginning, node A is in U-mode. No feedback is sent by node B until the acknowledgment sent by B changes the operation mode of compressor. The ROHC negotiation is made when the channel is open between A and B, then the NCP v6 packet is sent to node B. Then, the PPP interface starts sending the packets. The values of the different parameters are changed before a simulation because they are not negotiated by the actual ROHC standard. We have developed a UMTS error simulator, which generates UMTS error traces. The error traces are first generated as random sequences based on the error levels of the UMTS radio link [7]. These error traces are fed to our IPv6 ROHC Profile 1 implementation. In the graph of Figure 5, we can see the general behavior of ROHC in every operation modes. The size of original header of IPv6 is 40 bytes shown as a dashed line in Figure 5. The solid line indicates the size of the

compressed headers. The compressor always starts in the U-mode and the bigger headers seen in Figure 5 are the IR packets, which are sent periodically in the U-mode to refresh static information in the context of the decompressor.

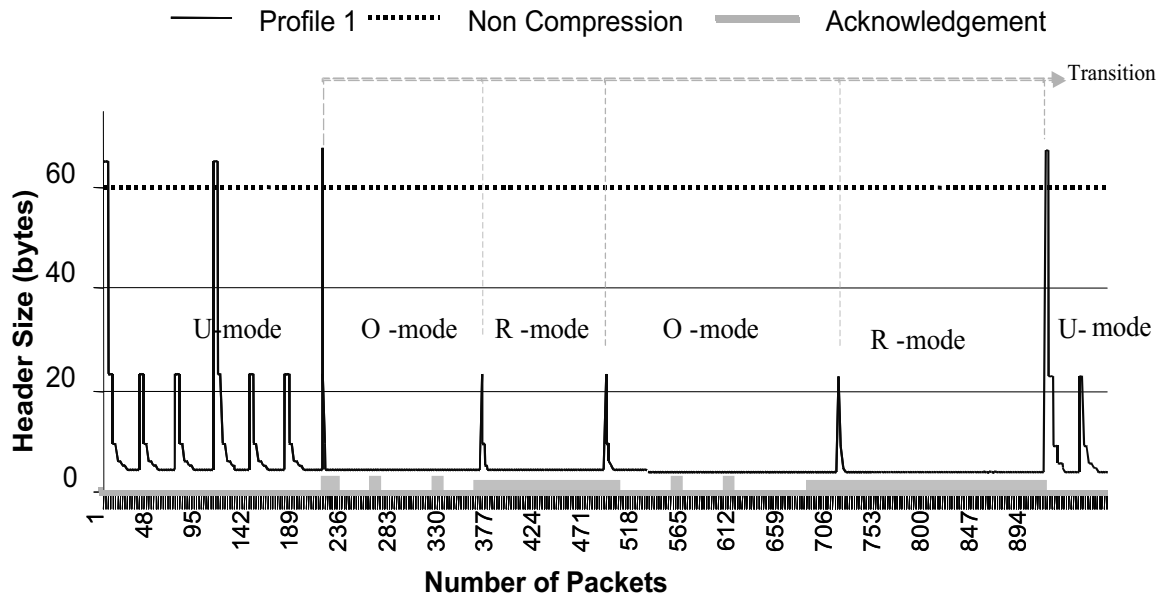


Figure 5: ROHC compression. Illustration of Operation Modes (U, O, R), Transition Modes and the use of feedback with small error.

The mode transitions can also be seen in Figure 5 and it can be noted that while mode transition, the compressor cannot send the smallest compressed headers that are of SO compression level. We can also see the difference between the three modes and compare their compression efficiency. We notice the difference between O-mode and R-mode and see the increased use of feedback channel in the later. The R-mode is interesting because the confidence system (L) is not used as in U and O modes. The larger headers are only sent at

the beginning or when the context is lost and most of the time, SO packets are used. The performance in the uplink is improved in R-mode because the smallest headers are used more frequently than the larger ones. It is important to mention that Figure 5 is the evolution of the compression with small BER and it is used only for comparing compression efficiency. When the error is small, the O-mode performance is found better than the R-mode because downlink is used less in the former than the latter; but the case is entirely different in a noisy link where the increased use of the downlink in the R-mode gives better feedback and hence, better robustness. ROHC over a radio link for IPv6 gives two types of errors. They occur in two types of situations as shown in Figure 6. First is when an error occurs in the compressed header and the decompressor catches it by ROHC CRC. The decompressor then drops the packet, it does not forward it to the IPv6 layer, when this happens the error is produced by ROHC and we call it ROHC loss. The other situation is when an error occurs in the payload then the decompressor is unable to catch the error and forwards it to the IPv6 layer. When the payload is handed to the UDP and since for IPv6 the UDP checksum is mandatory, the error is caught and the packet is discarded, we call it Application loss. This loss can be undesired because most sources (video, speech coders) developed for cellular links tolerate errors in the encoded data. Such coders will want damaged packets to be still delivered to them and will not want to enable UDP Checksum. The solution would be the use of UDP-Lite but this implies for ROHC a new profile that is not still standardized. The performance of ROHC compression with different error rates is shown in Figure 7, we can see that the BER affects more often the payload than the header information. From Figure 7 we find that the Application loss as compared to the ROHC loss is significant and would lead to bad performance.

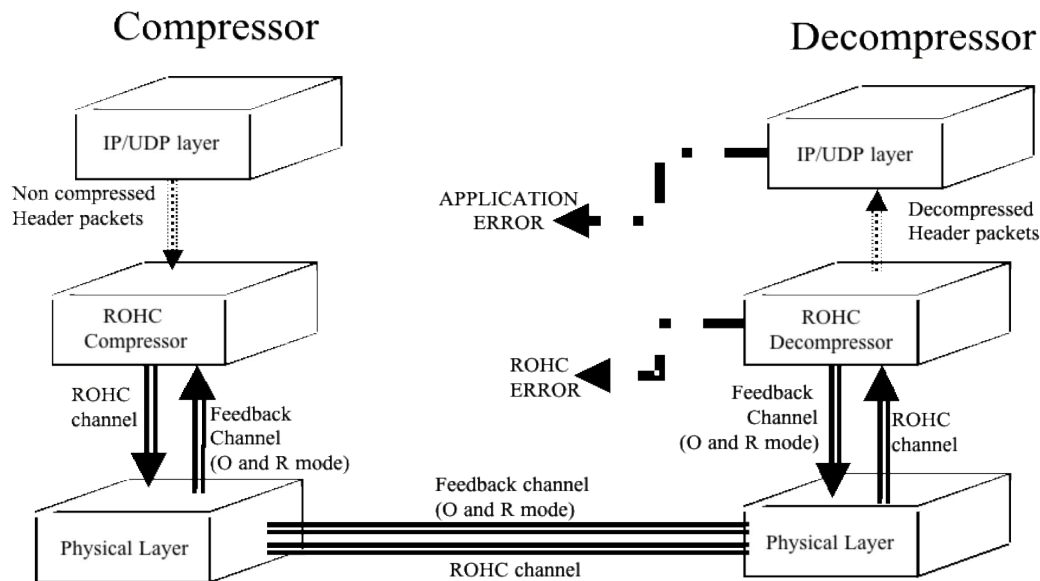


Figure 6: The ROHC Architecture has two types of error.

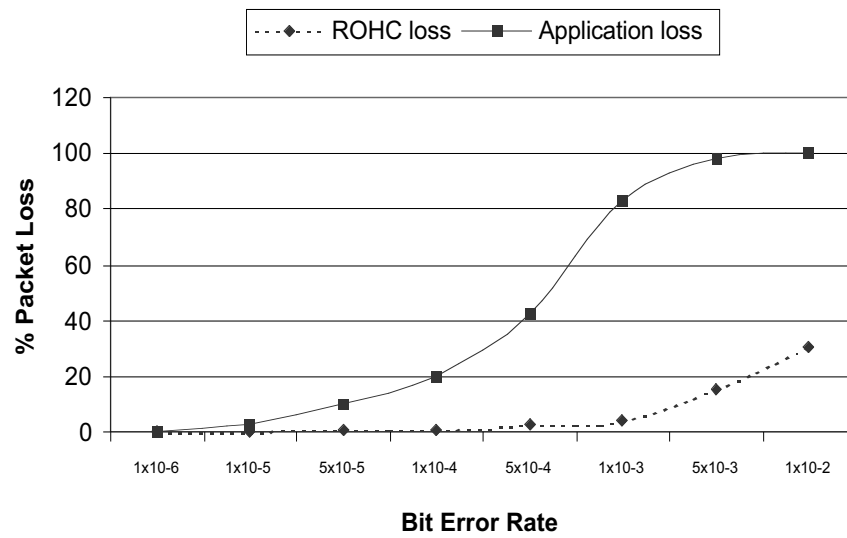


Figure 7: ROHC loss vs. application loss for different BER, when UDP checksum is enabled.

5 Analytical Model

We consider a packet flow which may represent the transfer of a large data file from a video streaming server to a client. The flow is composed of successive and identical cycles which correspond to the sending of T packets, where T is called the timer. Each cycle is itself composed of five compression phases. In each of the first four phases, L packets are sent with an increasing compression degree. We denote by $\ell(n)$ the compressed header length of a packet in the n -th phase. The L packets of the first phase are sent with their header uncompressed, which means that $\ell(1) = 67$ bytes. This first phase is the initialization phase of the compression process. The L packets headers of the second, third and fourth phases are compressed to $\ell(2) = 23$ bytes, $\ell(3) = 6$ bytes, and $\ell(4) = 5$ bytes respectively. In the fifth phase, the remaining $T - 4L$ packets of the cycle are sent with their headers compressed to $\ell(5) = 4$ bytes. Next, a new cycle with the same behavior begins and so on. Thus, the parameters T and L must be such that $T > 4L$. Their values have to be adjusted in order to satisfy performances criteria as we shall see in the examples.

It must be noted that when the L packets headers of the first phase are all erroneously transmitted then all the packets of the whole cycle will be lost because the protocol needs at least one uncompressed packet, i.e. one packet of the first phase, to be transmitted correctly.

We denote by b the Bit Error Rate (BER), that is the probability that a bit is not correctly received and by m the payload size of a packet, that is the quantity, in bits of useful data to be transmitted (of course the header is not included in the payload). At the n -th phase, $n = 1, 2, 3, 4, 5$, the header length $\ell(n)$, in bits, of a packet is composed of two parts: a part of this header having size $c(n)$, which may be corrected after some faults with probability p , using error correction heuristics and the rest of size $\ell(n) - c(n)$ which cannot be corrected.

The values of $c(n)$ in bits are $c(1) = 80$, $c(2) = 80$, $c(3) = 12$, $c(4) = 10$ and $c(5) = 4$. and we suppose that the events associated with the error transmissions are independent. The probability p_n of sending correctly a packet in the phase n is thus given, for $1 \leq n \leq 5$, by

$$p_n = (1 - b)^{m+\ell(n)} + (1 - b)^{m+\ell(n)-c(n)}(1 - (1 - b)^{c(n)})p.$$

The first term corresponds to the fact that the whole packet has been sent correctly and the second term corresponds to the fact that at least one bit of the part of length $c(n)$ has failed but it has been successfully corrected.

After the transmission of a packet, we will say that we had a *success* if the packet arrived correctly (either because there was no error, or because errors were all corrected), a *failure* if the transmission failed but *if, in the same cycle, not all the transmissions failed*, and a *total failure* if the transmission failed as well as all the previous transmissions in the cycle. Let us code 1, 2 and 3 these three possible transmission results, and let us also call them *levels*. We will define a stochastic process $\{X_n, n \geq 1\}$, where $X_n = (i, l)$ if after transmitting the n th packet the result is $l \in \{1, 2, 3\}$, i being the rank of the packet in its cycle, that is, $i = [(n - 1) \bmod T] + 1$. It is immediate to see that $\{X_n\}$ is a discrete time homogeneous Markov chain whose associated transition graph is depicted in Figure 8 for the case $L = 3$ and $T = 14$. The states of the first level represented using circles are denoted by $(i, 1)$, for

$i = 1, \dots, T$, from left to right. The states of the second level represented using squares are denoted by $(i, 2)$, for $i = 2, \dots, T$ (state $(1, 2)$ is not feasible), from left to right and the states of the third level in diamond shape are denoted by $(i, 3)$, for $i = 1, \dots, T$, from left to right.

Since we are interested in the stationary behavior of the protocol, we denote by X the stationary version of the Markov chain $\{X_n\}$. This version exists since the Markov chain is finite and irreducible.

The transition from state $(1, 1)$ to state $(2, 1)$ corresponds to the successful transmission of the second packet of the current cycle, so its probability is equal to p_1 , when $L > 1$. In the same way, a transition from state $(i - 1, 1)$ to state $(i, 1)$ or from state $(i - 1, 2)$ to state $(i, 1)$, for $i = 2, \dots, T$, corresponds to the successful transmission of the i -th packet of the current cycle, so its probability is equal to p_n if states $(i, 1)$ or $(i - 1, 2)$ belongs to the n -th phase of the cycle. The transition probability from state $(T, 1)$ to state $(1, 1)$ or from state $(T, 2)$ to state $(1, 1)$ is thus equal p_1 and means the successful transmission of the first packet of the next cycle.

The transition from state $(1, 1)$ to state $(2, 2)$ corresponds to the unsuccessful transmission of the second packet of the current cycle, so its probability is equal to $1 - p_1$, when $L > 1$. In the same way, a transition from state $(i - 1, 1)$ to state $(i, 2)$ or from state $(i - 1, 2)$ to state $(i, 2)$, for $i = 3, \dots, T$, corresponds to the unsuccessful transmission of the i -th packet of the current cycle, so its probability is equal to $1 - p_n$ if state $(i - 1, 1)$ belongs to the n -th phase of the cycle. The transition probability from state $(T, 1)$ to state $(1, 3)$ or from state $(T, 2)$ to state $(1, 3)$ is thus equal $1 - p_1$ and means the unsuccessful transmission of the first packet of the next cycle. A new cycle then begins with the same probabilistic behavior.

The states of the third level (diamond shape) that we denoted by $(i, 3)$, for $i = 1, \dots, T$, from the left to the right are used to represent the case where the L packets headers of the first phase are all erroneously transmitted which leads to the lost of all the packets of the whole cycle.

The state $(1, 3)$ of a current cycle is reached when the first packet of this cycle is unsuccessfully transmitted, i.e. from states $(T, 1)$, $(T, 2)$ or $(T, 3)$ with the same probability $1 - p_1$. Next, a transition from state $(i - 1, 3)$ to state $(i, 3)$, for $i = 2, \dots, L$, corresponds to the unsuccessful transmission of the i -th packet, so its probability is equal to $1 - p_1$. In the same way, a transition from state $(i - 1, 3)$ to state $(i, 1)$, for $i = 2, \dots, L$, corresponds to the successful transmission of the i -th packet, so its probability is equal to p_1 . When the process is in state $(L, 3)$, this means that the L first packets of the cycle have been unsuccessfully transmitted, so the transition probability from state $(i, 3)$ to state $(T, 3)$, for $i = L, \dots, T$, is equal to 1.

We partition the state space in a vertical way (see Figure 8) into T subsets as follows. For $i = 1, \dots, T$, we introduce the subset of states S_i defined by

$$S_1 = \{(1, 1), (1, 3)\} \text{ and } S_i = \{(i, 1), (i, 2), (i, 3)\} \text{ for } i = 2, \dots, T.$$

Using this partition, the transition probability matrix P of the Markov chain $\{X_n\}$ is given by the submatrices $P_{i,j}$ containing the transition probabilities from states of S_i to states of

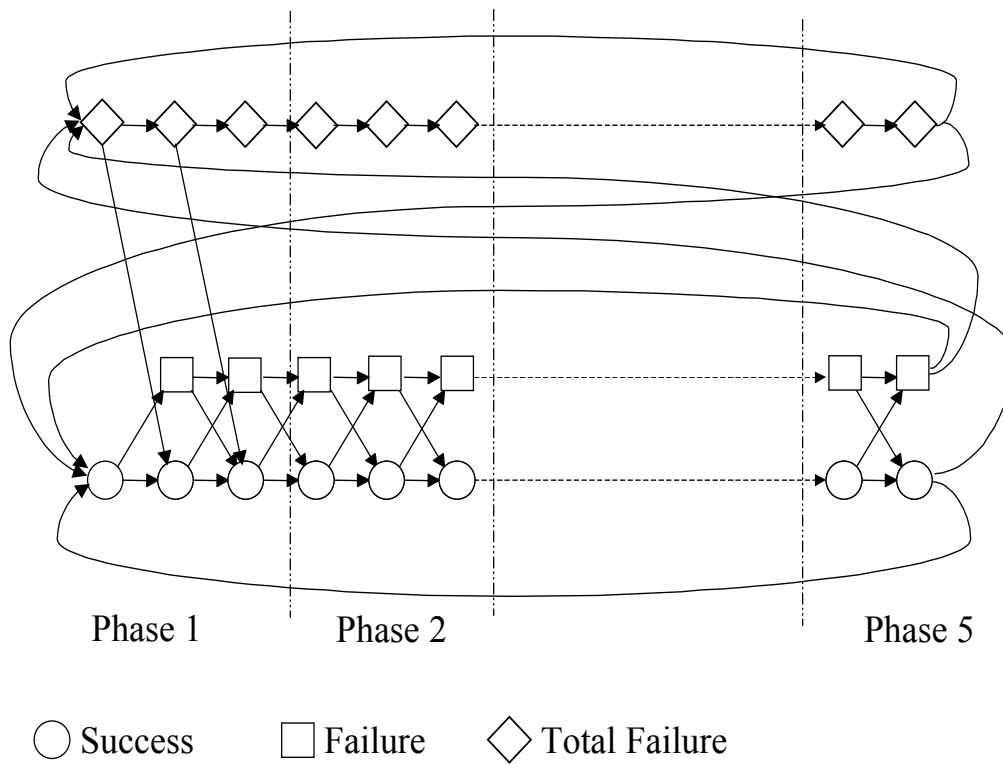


Figure 8: The Markov chain.

S_j . The non-zero submatrices of matrix P are given by

$$P_{i,i+1} = \begin{cases} A & \text{for } i = 1 \\ B_1 & \text{for } 2 \leq i \leq L - 1 \\ B_2 & \text{for } L \leq i \leq 2L - 1 \\ B_3 & \text{for } 2L \leq i \leq 3L - 1 \\ B_4 & \text{for } 3L \leq i \leq 4L - 1 \\ B_5 & \text{for } 4L \leq i \leq T - 1 \end{cases} \quad \text{and } P_{T,1} = C,$$

where the matrices A , B_n and C are defined by

$$A = \begin{pmatrix} p_1 & 1-p_1 & 0 \\ p_1 & 0 & 1-p_1 \end{pmatrix}; \quad B_1 = \begin{pmatrix} p_1 & 1-p_1 & 0 \\ p_1 & 1-p_1 & 0 \\ p_1 & 0 & 1-p_1 \end{pmatrix};$$

$$B_n = \begin{pmatrix} p_n & 1-p_n & 0 \\ p_n & 1-p_n & 0 \\ 0 & 0 & 1 \end{pmatrix}, \quad \text{for } 2 \leq n \leq 5 \quad \text{and } C = \begin{pmatrix} p_1 & 1-p_1 \\ p_1 & 1-p_1 \\ p_1 & 1-p_1 \end{pmatrix}.$$

For example, in the case depicted in Figure 8, where $L = 3$ and $T = 14$, we have

$$P = \left(\begin{array}{ccc|ccc|ccc|ccc|cc} 0 & A & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 \\ 0 & 0 & B_1 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 \\ 0 & 0 & 0 & B_2 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 \\ \hline 0 & 0 & 0 & 0 & B_2 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & B_2 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & 0 & B_3 & 0 & 0 & 0 & 0 & 0 & 0 & 0 \\ \hline 0 & 0 & 0 & 0 & 0 & 0 & 0 & B_3 & 0 & 0 & 0 & 0 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & B_3 & 0 & 0 & 0 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & B_4 & 0 & 0 & 0 & 0 \\ \hline 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & B_4 & 0 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & B_4 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & B_5 & 0 \\ \hline 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & B_5 \\ C & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 & 0 \end{array} \right).$$

We denote by π the stationary distribution of the Markov chain. We thus have

$$\pi = \pi P \quad \text{and} \quad \pi \mathbb{1} = 1, \tag{1}$$

where $\mathbb{1}$ denotes the column vector with all entries equal to 1, its dimension being specified by the context where it is used. We decompose the vector π using the partition of the state space defined by the subsets S_i . So, for $i = 1, \dots, T$, we define the subvectors π_i such that

$$\pi = (\pi_1, \dots, \pi_L, \pi_{L+1}, \dots, \pi_{2L}, \pi_{2L+1}, \dots, \pi_{3L}, \pi_{3L+1}, \dots, \pi_{4L}, \pi_{4L+1}, \dots, \pi_T),$$

where π_i is the row subvector of vector π containing the entries corresponding to subset S_i , that is

$$\pi_1 = (\pi(1, 1), \pi(1, 3)) \text{ and } \pi_i = (\pi(i, 1), \pi(i, 2), \pi(i, 3)) \text{ for } i = 2, \dots, T.$$

The linear system (1) then becomes in terms of the subvectors π_i and submatrices A , B_n , for $n = 1, 2, 3, 4, 5$ and C

$$\left\{ \begin{array}{l} \pi_1 = \pi_T C \\ \pi_2 = \pi_1 A \\ \pi_3 = \pi_2 B_1 \\ \vdots \\ \pi_L = \pi_{L-1} B_1 \end{array} \right. ; \left\{ \begin{array}{l} \pi_{L+1} = \pi_L B_2 \\ \pi_{L+2} = \pi_{L+1} B_2 \\ \pi_{L+3} = \pi_{L+2} B_2 \\ \vdots \\ \pi_{2L} = \pi_{2L-1} B_2 \end{array} \right. ; \left\{ \begin{array}{l} \pi_{2L+1} = \pi_{2L} B_3 \\ \pi_{2L+2} = \pi_{2L+1} B_3 \\ \pi_{2L+3} = \pi_{2L+2} B_3 \\ \vdots \\ \pi_{3L} = \pi_{3L-1} B_3 \end{array} \right. ; \quad (2)$$

$$\left\{ \begin{array}{l} \pi_{3L+1} = \pi_{3L} B_4 \\ \pi_{3L+2} = \pi_{3L+1} B_4 \\ \pi_{3L+3} = \pi_{3L+2} B_4 \\ \vdots \\ \pi_{4L} = \pi_{4L-1} B_4 \end{array} \right. ; \left\{ \begin{array}{l} \pi_{4L+1} = \pi_{4L} B_5 \\ \pi_{4L+2} = \pi_{4L+1} B_5 \\ \pi_{4L+3} = \pi_{4L+2} B_5 \\ \vdots \\ \pi_T = \pi_{T-1} B_5 \end{array} \right.$$

The submatrices A , B_n , for $n = 1, 2, 3, 4, 5$ and C are stochastic, so by multiplying the equations of linear system (2) by the column vector $\mathbb{1}$ on the right, we get

$$\pi_1 \mathbb{1} = \pi_2 \mathbb{1} = \dots = \pi_T \mathbb{1},$$

which leads, by using the equation $\pi \mathbb{1} = 1$, to

$$\pi_i \mathbb{1} = \frac{1}{T}, \text{ for } 1 \leq i \leq T.$$

It is now very easy to solve system (2). The first set of equations gives us

$$\pi_1 = \frac{1}{T} (p_1, 1 - p_1).$$

For $2 \leq i \leq L$,

$$\pi_i = \frac{1}{T} (p_1, 1 - p_1 - (1 - p_1)^i, (1 - p_1)^i).$$

The four other sets of equations give us:

For $L + 1 \leq i \leq 2L$,

$$\pi_i = \frac{1}{T} (p_2(1 - (1 - p_1)^L), (1 - p_2)(1 - (1 - p_1)^L), (1 - p_1)^L).$$

For $2L + 1 \leq i \leq 3L$

$$\pi_i = \frac{1}{T} (p_3(1 - (1 - p_1)^L), (1 - p_3)(1 - (1 - p_1)^L), (1 - p_1)^L).$$

For $3L + 1 \leq i \leq 4L$

$$\pi_i = \frac{1}{T} (p_4(1 - (1 - p_1)^L), (1 - p_4)(1 - (1 - p_1)^L), (1 - p_1)^L).$$

For $4L + 1 \leq i \leq T$

$$\pi_i = \frac{1}{T} (p_5(1 - (1 - p_1)^L), (1 - p_5)(1 - (1 - p_1)^L), (1 - p_1)^L).$$

We denote by p_{hit} the probability that the transmission of all packets is correct. It is thus the probability for the Markov chain to be in one of the states $(i, 1)$, for $i = 1, \dots, T$. This probability is also the sum of the first entries of all subvectors π_i . Thus we have

$$p_{\text{hit}} = \sum_{i=1}^T \pi(i, 1) = \sum_{i=1}^T \pi_i(1),$$

which leads to

$$p_{\text{hit}} = \frac{1}{T} \left[Lp_1 + \left(L(p_2 + p_3 + p_4) + (T - 4L)p_5 \right) \left(1 - (1 - p_1)^L \right) \right]. \quad (3)$$

6 Numerical Results

In order to validate our analytical model we compared its results with the results obtained by the implementation and we observed that the accuracy is quite good. Since we focused on ROHC effects, we made our experiments without payloads, i. e. with $m = 0$. The values of the other input parameters are the following: the timer has been fixed to $T = 800$ and the probability p associated with the error corrections is $p = 0$. To illustrate this, Figure 9 represents the stationary packet loss probability p_{loss} , which is equal to $1 - p_{\text{hit}}$, as a function of the phases length L for different values of the bit error rate b . This figure shows that our analytical model is accurate since the maximal difference between the implementation and the analytical model is usually less than 10%.

Using our model, we can compute the stationary packet loss probability for several parameter values.

In Figure 10, we show the loss probability as function of L for different values of T , when $b = 0.01$ and $p = 0$. For every value of T , the packet loss probability is very high, which means that when $b = 0.01$ we have to increase the length L of compression phases. Note that T must increase too since we must have $T > 4L$. We ran the model for $L = 100$ and we got the loss probability equal to 0.82 for $T = 600$ and 0.8 for $T = 800$ which are again too large loss probabilities. Note that from relation (3), when L is large, the loss probability p_{loss} , is approximatively proportional to the ratio L/T . More formally, when $L \rightarrow \infty$ and $T \rightarrow \infty$ and $L/T \rightarrow r$, with $r < 1/4$, we have

$$p_{\text{loss}} \rightarrow r(p_1 + p_2 + p_3 + p_4) + (1 - 4r)p_5.$$

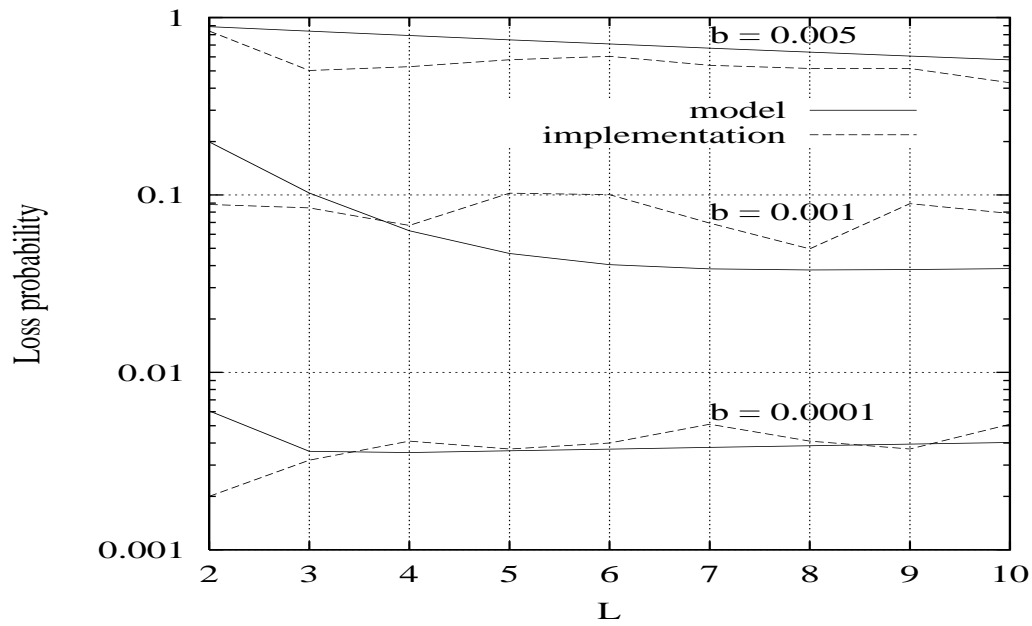


Figure 9: Validation of the analytical model.

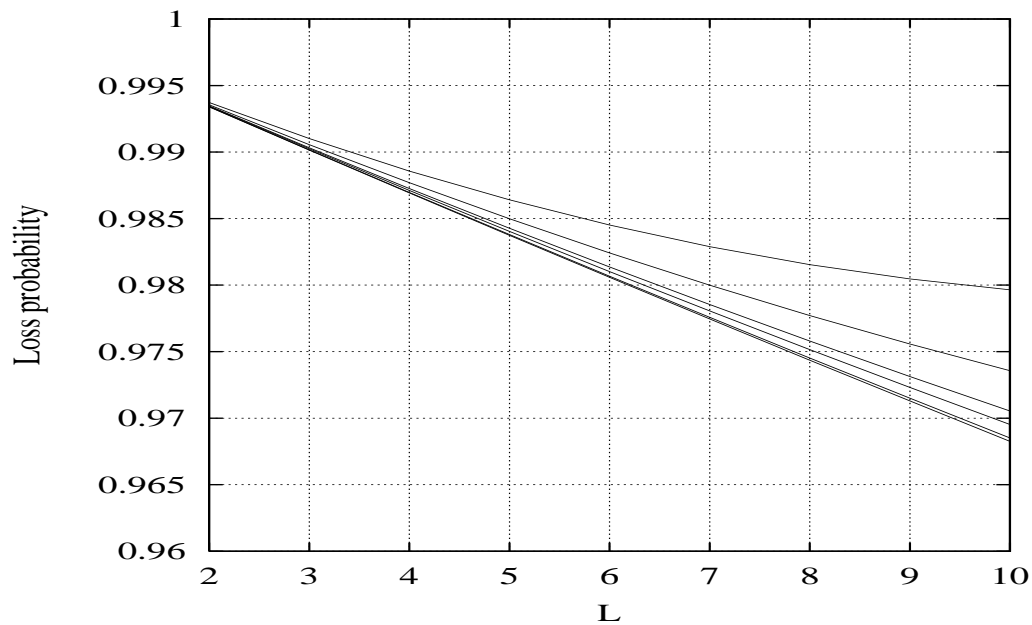
Figure 10: For $b = 0.01$ from top to bottom: $T = 50$, $T = 100$, $T = 200$, $T = 300$, $T = 600$, $T = 800$.

Figure 11 and 12 also show the behavior of the packet loss probability for $b = 10^{-3}$ and $b = 10^{-4}$ respectively. These figures allow us to choose the parameters b , L and T so that the loss probability is less than or equal to a given value.

For instance, when $b = 10^{-3}$, Figure 11 shows that the value $L = 5$ corresponds to the lowest loss probability, $p_{\text{loss}} = 0.096$ for $T = 50$. This figure also shows that the global minimum loss probability is reached when $L = 8$ and $T = 800$, where we have $p_{\text{loss}} = 0.0378$.

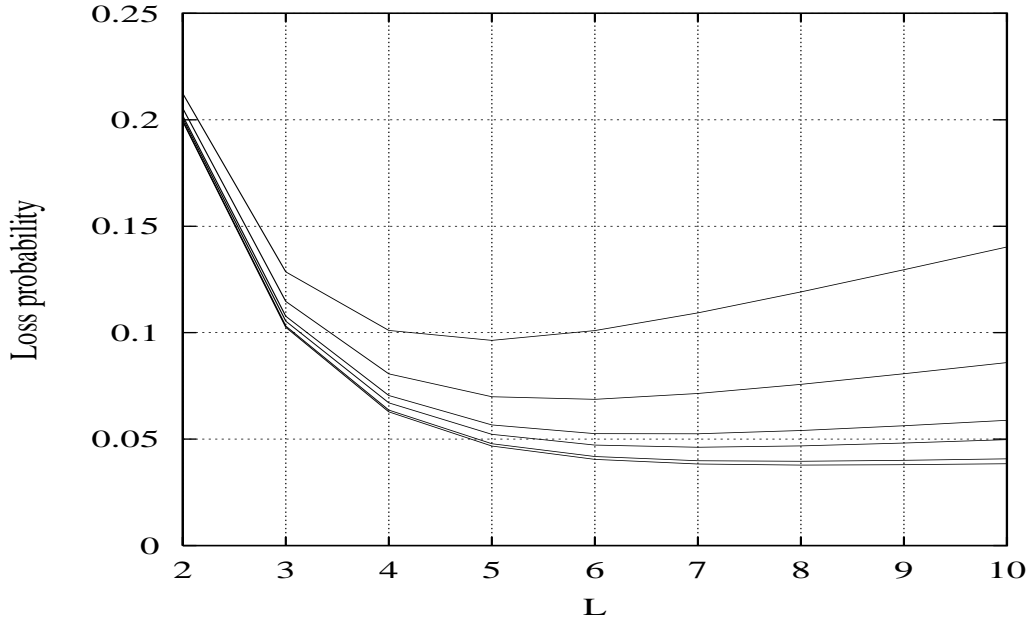


Figure 11: For $b = 0.001$ from top to bottom: $T = 50$, $T = 100$, $T = 200$, $T = 300$, $T = 600$, $T = 800$.

When $b = 10^{-4}$, Figure 12 shows that the global minimum loss probability is reached when $L = 4$ and $T = 800$ where we have $p_{\text{loss}} = 0.0035$.

7 Conclusion

The ROHC description of this paper gives an overview of the protocol complexity and the number of different parameters that must be set. We developed an analytical model of this protocol based only on the first compressor profile with the unidirectional mode. This mode represents the most common use of ROHC in low bandwidth links. We have shown that this model, described by a Markov chain, is quite accurate and, thus, that the analytical results it provides constitute a step towards establishing relations between configuration parameters and the error rate. This will help the configuration process itself, will contribute to the auto-configuration of the link layer and will allow to optimize the performance of the header compression algorithm.

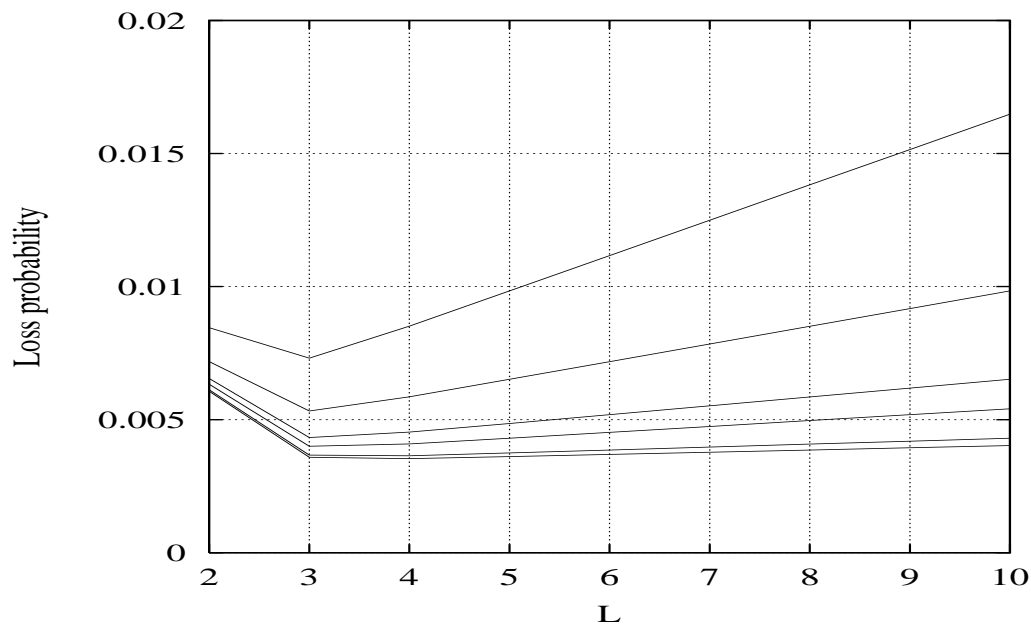


Figure 12: For $b = 0.0001$ from top to bottom: $T = 50$, $T = 100$, $T = 200$, $T = 300$, $T = 600$, $T = 800$.

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Unité de recherche INRIA Lorraine, Technopôle de Nancy-Brabois, Campus scientifique,
615 rue du Jardin Botanique, BP 101, 54600 VILLERS LÈS NANCY
Unité de recherche INRIA Rennes, Irisa, Campus universitaire de Beaulieu, 35042 RENNES Cedex
Unité de recherche INRIA Rhône-Alpes, 655, avenue de l'Europe, 38330 MONTBONNOT ST MARTIN
Unité de recherche INRIA Rocquencourt, Domaine de Voluceau, Rocquencourt, BP 105, 78153 LE CHESNAY Cedex
Unité de recherche INRIA Sophia-Antipolis, 2004 route des Lucioles, BP 93, 06902 SOPHIA-ANTIPOLIS Cedex

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