

The Multi-Class Extension to Multi-Link PPP
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Abstract

A companion document describes an architecture for providing integrated services over low-bitrate links, such as modem lines, ISDN B-channels, and sub-T1 links [[1](#)]. The main components of the architecture are: a real-time encapsulation format for asynchronous and synchronous low-bitrate links, a header compression architecture optimized for real-time flows, elements of negotiation protocols used between routers (or between hosts and routers), and announcement protocols used by applications to allow this negotiation to take place.

This document proposes the fragment-oriented solution for the real-time encapsulation format part of the architecture. The general approach is to start from the PPP Multilink fragmentation protocol [[2](#)] and provide a small number of extensions to add functionality and reduce the overhead.

This document is a product of the IETF ISSLL working group. Comments are solicited and should be addressed to the two working groups' mailing lists at [issll@mercury.lcs.mit.edu](#) and [ietf-](#)

1. Introduction

As an extension to the ``best-effort'' services the Internet is well-known for, additional types of services (``integrated services'') that support the transport of real-time multimedia information are being developed for and deployed in the Internet.

A companion document describes an architecture for providing integrated services over low-bitrate links, such as modem lines, ISDN B-channels, and sub-T1 links [1]. The main components of the architecture are: a real-time encapsulation format for asynchronous and synchronous low-bitrate links, a header compression architecture optimized for real-time flows, elements of negotiation protocols used between routers (or between hosts and routers), and announcement protocols used by applications to allow this negotiation to take place.

The present document defines the fragment-oriented solution for the real-time encapsulation format part of the architecture, i.e. for the queues-of-fragments type sender [1]. As described in more detail in the architecture document, a real-time encapsulation format is required as, e.g., a 1500 byte packet on a 28.8 kbit/s modem link makes this link unavailable for the transmission of real-time information for about 400 ms. This adds a worst-case delay that causes real-time applications to operate with round-trip delays on the order of at least a second -- unacceptable for real-time conversation. The PPP extensions defined in this document allow a sender to fragment the packets of various priorities into multiple classes of fragments, allowing high-priority packets to be sent between fragments of lower priorities.

A companion document based on these extensions [5] defines a suspend/resume-oriented solution for those cases where the best possible delay is required and the senders are of type 1 [1].

2. Requirements

The main design goal for the components of an architecture that addresses real-time multimedia flows over low-bitrate links is that of minimizing the end-to-end delay. More specifically, the worst

case delay (after removing possible outliers, which are equivalent to packet losses from an application point of view) is what determines the playout points selected by the applications and thus the delay actually perceived by the user.

In addition, every attempt should obviously be undertaken to maximize the bandwidth actually available to media data; overheads must be minimized.

The solution should not place unnecessary burdens on the non-real-time flows. In particular, the usual MTU should be available to these flows.

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The most general approach would provide the ability to suspend any packet (real-time or not) for a more urgent real-time packet, up to an infinite number of levels of nesting. On the other hand, it is likely that there would rarely be a requirement for a real-time packet to suspend another real-time packet that is not at least about twice as long. Typically, the largest packet size to be expected on a PPP link is the default MTU of 1500 bytes. The smallest high-priority packets are likely to have on the order of 22 bytes (compressed RTP/G.723.1 packets). In the 1:72 range of packet sizes to be expected, this translates to a maximum requirement of about eight levels of suspension (including one level where long real-time packets suspend long non-real-time packets). On 28.8kbit/s modems, there seems to be a practical requirement for at least two levels of suspension (i.e., audio suspends any longer packet including video, video suspends other very long packets).

On an architectural level, there are several additional requirements for the fragmentation scheme:

- a) The scheme must be predictable enough that admission control can make decisions based on its characteristics. As is argued in [\[1\]](#), this will often only be the case when additional hints about the characteristics of the flow itself are available (application hints).
- b) The scheme must be robust against errors, at least with the same level of error detection as PPP.
- c) The scheme must in general cooperate nicely with PPP. In particular, it should be as compatible to existing PPP standards as possible. On a link that (based on PPP negotiation) makes use of the scheme, it should always be possible to fall back to

standard LCP without ambiguity.

- d) The scheme must work well with existing chips and router systems. (See [1] for a more extensive discussion of implementation models.) For synchronous links this means using HDLC framing; with much existing hardware, it is also hard to switch off the HDLC per-frame CRC. For asynchronous links, there is much more freedom in design; on the other hand, a design that treats them much different from synchronous links would lose a number of desirable properties of PPP.
- e) The scheme must be future proof. In particular, the emergence of V.80 based modems may significantly change the way PPP is used with modems.

This document does not address additional requirements that may be relevant in conjunction with Frame Relay; however, there seems to be little problem in applying the principles of this document to ``PPP in Frame Relay'' [3].

[3.](#) Using PPP Multilink as-is

Transmitting only part of a packet to allow higher-priority traffic to intervene and resuming its transmission later on is a kind of fragmentation. The existing PPP Multilink Protocol (MP, [2]) provides for sequence numbering and begin/end bits, allowing packets to be split into fragments (Figure 1).

Figure 1: Multilink Short Sequence Number Fragment Format [2]

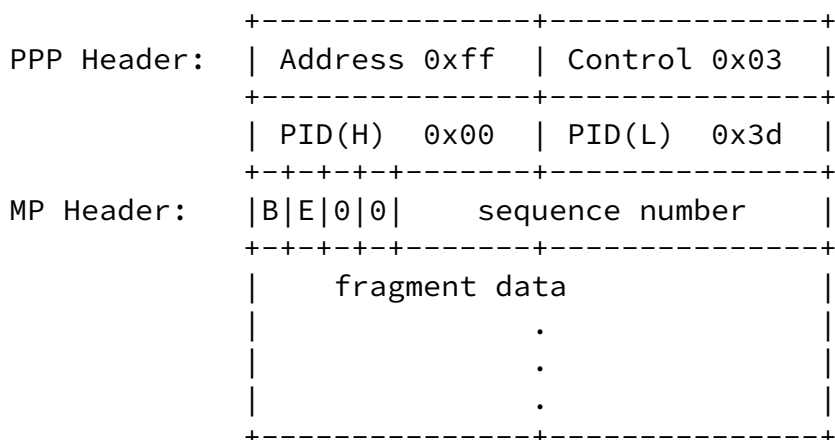
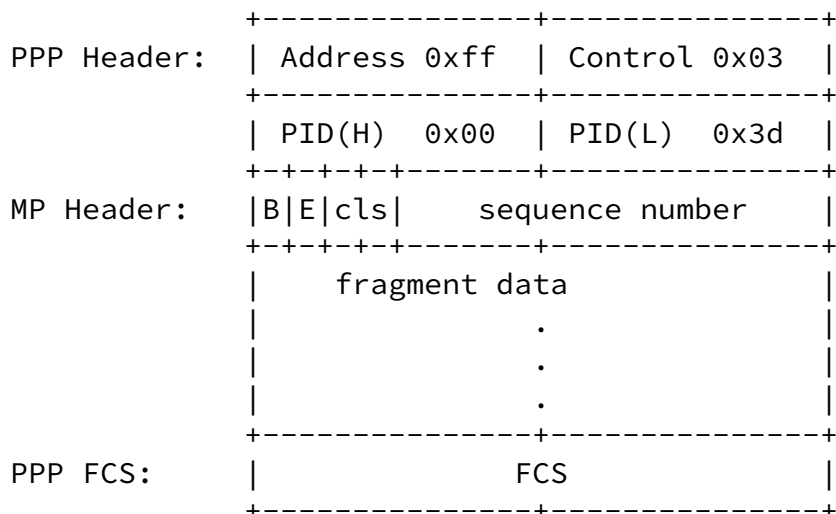


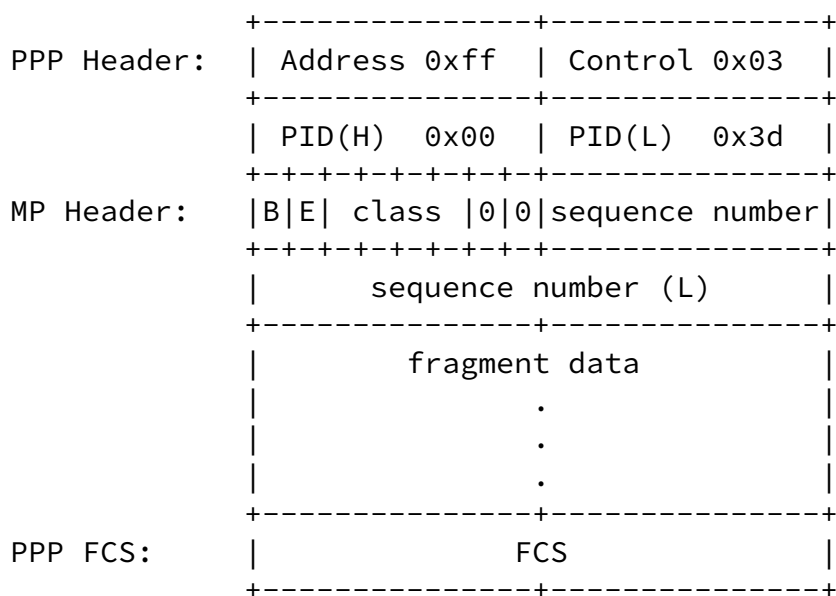
Figure 2: Short Sequence Number Fragment Format With Classes



Each class runs a separate copy of the mechanism defined in [2], i.e. uses a separate sequence number space and reassembly buffer.

Similarly, for the long sequence number format:

Figure 3: Long Sequence Number Fragment Format With Classes



Together with the ability to send packets without a multilink header, this provides four levels of suspension with 12-bit headers (probably sufficient for many practical applications) and sixteen levels with 24-bit headers (only four of the six free bits are used in this case -- based on the rationale given above, sixteen levels should

generally be more than sufficient).

[5.](#) Prefix elision: Compressing common header bytes

For some applications, all packets of a certain class will have a common protocol identifier (or even more than one common prefix byte). In this case, the following optimization is possible: the class number can be associated with a prefix of bytes that are removed from each packet before transmission and that are implicitly prepended to the reassembled packet after reception.

Note that if only some of the packets to be transmitted at a certain level of priority have the common prefix, it may still be possible to utilize this method by allocating two class numbers and only associating one of them with the prefix. (This is the reason why four of the unused bits in the long sequence number format have been allocated to the class number instead of the three that generally should suffice.)

Prefix elision is not a replacement for header compression or data compression: it allows to compress away prefixes that often are not reachable by these other methods.

[6.](#) Negotiable options

The following PPP LCP options are already defined by MP:

- o Multilink Maximum Received Reconstructed Unit
- o Multilink Short Sequence Number Header Format
- o Endpoint Discriminator

This document defines two new LCP options:

- o Multilink Header Format
- o Prefix Elision

[6.1.](#) Multilink header format option

A summary of the Multilink Header Format Option format is shown below. The fields are transmitted from left to right.

0										1										2									
0	1	2	3	4	5	6	7	8	9	0	1	2	3	4	5	6	7	8	9	0	1	2	3						
+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-						
Type = TBD										Length = 3										Code									
+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-	+	-						

The values defined for the use of this option are:

- An implementation **MUST NOT** request a combination of both the Short Sequence Number Header Format Option and this option.

This LCP option advises the peer that the implementation wishes to send only packets with a certain prefix in each of the given classes; this prefix is not sent as part of the information in the fragment(s) of this class. By default, this common prefix is empty for all classes. When this option is received, an implementation **MUST** either add the prefix given for the class to all subsequently received multilink packets of each of the given classes or configure-NAK or configure-Reject the option.

- [3] W. Simpson, ``PPP in Frame Relay'', [RFC 1973](#), June 1996.
- [4] R. Andrades, F. Burg, ``QOSPPP Framing Extensions to PPP'', Work in Progress ([draft-andrades-framing-ext-00.txt](#)), September 1996.
- [5] C. Bormann, ``PPP in a real-time oriented HDLC-like framing'', Work in Progress ([draft-ietf-issll-isslow-rtf-01.txt](#)), May 1997.

[9.](#) Addresses

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[9.1.](#) Working Group

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