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A. Clark
Telchemy
S. Zhang, Ed.
J. Zhao
STTRI
Q. Wu
Huawei
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**RTP Control Protocol (RTCP) Extended Report (XR) Block for Burst/Gap
Loss metric Reporting
draft-ietf-xrblock-rtcp-xr-burst-gap-loss-07.txt**

Abstract

This document defines an RTP Control Protocol (RTCP) Extended Report (XR) Block that allows the reporting of Burst and Gap Loss metrics for use in a range of RTP applications.

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Table of Contents

1.	Introduction	3
1.1.	Burst and Gap Loss Report Block	3
1.2.	RTCP and RTCP XR Reports	3
1.3.	Performance Metrics Framework	3
1.4.	Applicability	4
2.	Terminology	5
2.1.	Standards Language	5
3.	Burst/Gap Loss Block	6
3.1.	Report Block Structure	6
3.2.	Definition of Fields in Burst/Gap Loss Report Block	6
3.3.	Derived metrics based on reported metrics	9
4.	Considerations for Voice-over-IP applications	11
5.	SDP Signaling	12
5.1.	SDP rtcp-xr-attrib Attribute Extension	12
5.2.	Offer/Answer Usage	12
6.	IANA Considerations	13
6.1.	New RTCP XR Block Type value	13
6.2.	New RTCP XR SDP Parameter	13
6.3.	Contact information for registrations	13
7.	Security Considerations	14
8.	Contributors	15
9.	Acknowledgments	16
10.	References	17
10.1.	Normative References	17
10.2.	Informative References	17
Appendix A.	Change Log	18
A.1.	draft-ietf-xrblock-rtcp-xr-burst-gap-loss-07	18
A.2.	draft-ietf-xrblock-rtcp-xr-burst-gap-loss-06	18
A.3.	draft-ietf-xrblock-rtcp-xr-burst-gap-loss-04	18
	Authors' Addresses	19

1. Introduction

1.1. Burst and Gap Loss Report Block

This draft defines a new block type to augment those defined in [\[RFC3611\]](#) for use in a range of RTP applications. The new block type supports the reporting of the proportion of packets lost by the network. The losses during loss bursts are reported, together with the number of bursts and additional data allowing the calculation of statistical parameters (mean and variance) of the distribution of burst lengths. Some uses of these metrics depend on the availability of the metric "cumulative number of packets lost" from RTCP [\[RFC3550\]](#).

This block provides information on transient IP problems. Burst/Gap metrics are typically used in Cumulative reports, however they also MAY be used in Interval reports. The burstiness of packet loss affects user experience, may influence any sender strategies to mitigate the problem, and may also have diagnostic value.

The metric belongs to the class of transport-related end system metrics defined in [\[RFC6792\]](#).

The definitions of Burst, Gap, Loss and Discard are consistent with definitions in [\[RFC3611\]](#). To accommodate the range of jitter buffer algorithms and packet discard logic that may be used by implementors, the method used to distinguish between bursts and gaps may be an equivalent method to that defined in [\[RFC3611\]](#). The method used SHOULD produce the same result as that defined in [\[RFC3611\]](#) for conditions of burst packet loss, but MAY produce different results for conditions of time varying jitter.

1.2. RTCP and RTCP XR Reports

The use of RTCP for reporting is defined in [\[RFC3550\]](#). [\[RFC3611\]](#) defined an extensible structure for reporting using an RTCP Extended Report (XR). This document defines a new Extended Report block for use with [\[RFC3550\]](#) and [\[RFC3611\]](#).

1.3. Performance Metrics Framework

The Performance Metrics Framework [\[RFC6390\]](#) provides guidance on the definition and specification of performance metrics. The RTP Monitoring Architectures [\[RFC6792\]](#) provides guideline for reporting block format using RTCP XR. The Metrics Block described in this document are in accordance with the guidelines in [\[RFC6390\]](#) and [\[RFC6792\]](#).

1.4. Applicability

These metrics are applicable to a range of RTP applications which contain jitter buffers and don't use stream repair means, e.g., Forward Error Correction (FEC) [[RFC5109](#)] and/or retransmission [[RFC4588](#)].

2. Terminology

2.1. Standards Language

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC 2119](#) [[RFC2119](#)].

In addition, the following terms are defined:

Received, Lost and Discarded

A packet shall be regarded as lost if it fails to arrive within an implementation-specific time window. A packet that arrives within this time window but is too early or late to be played out or thrown away before playout due to packet duplication or redundancy shall be regarded as discarded. A packet shall be classified as one of received (or OK), discarded or lost. The metric "cumulative number of packets lost" defined in [[RFC3550](#)] reports a count of packets lost from the media stream (single SSRC within single RTP session). Similarly the metric "number of packets discarded" defined in [[DISCARD](#)] reports a count of packets discarded from the media stream (single SSRC within single RTP session) arriving at the receiver. Another metric defined in [[RFC5725](#)] is available to report on packets which are not recovered by any repair techniques which may be in use.

Bursts and Gaps

The terms Burst and Gap are used in a manner consistent with that of RTCP XR [[RFC3611](#)]. RTCP XR views a RTP stream as being divided into bursts, which are periods during which the loss rate is high enough to cause noticeable quality degradation (generally over 5 percent loss rate), and gaps, which are periods during which lost packets are infrequent and hence quality is generally acceptable.

3. Burst/Gap Loss Block

Metrics in this block report on Burst/Gap Loss in the stream arriving at the RTP system. The measurement of these metrics are made at the receiving end of the RTP stream. Instances of this Metrics Block refer by Synchronization source (SSRC) to the separate auxiliary Measurement Information block [[RFC6776](#)] which describes measurement periods in use(see [RFC6776 section 4.2](#)). This Metrics Block relies on the measurement period in the Measurement Information block indicating the span of the report and SHOULD be sent in the same compound RTCP packet as the measurement information block. If the measurement period is not received in the same compound RTCP packet as this Metrics Block, this Metrics Block MUST be discarded.

3.1. Report Block Structure

Burst/Gap Loss metrics block

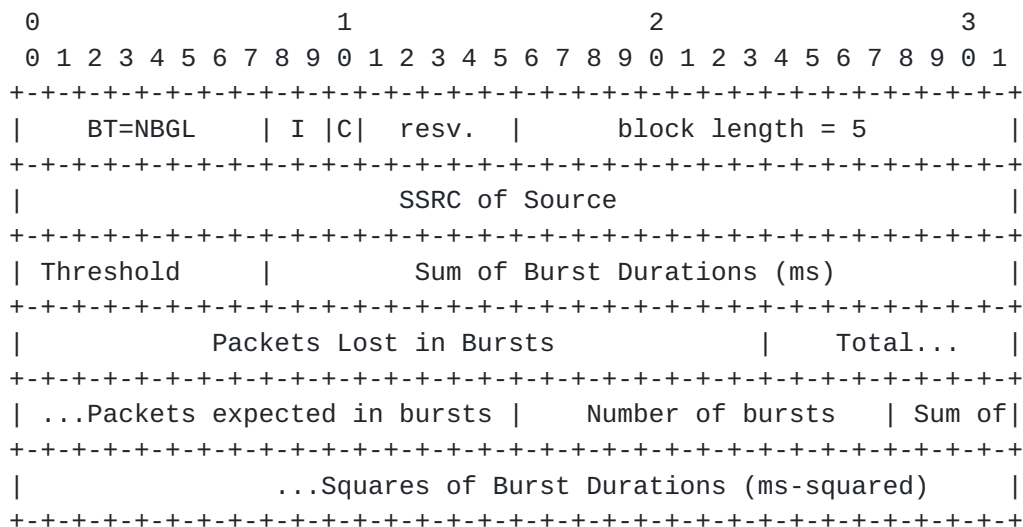


Figure 1: Report Block Structure

3.2. Definition of Fields in Burst/Gap Loss Report Block

Block type (BT): 8 bits

A Burst/Gap Loss Report Block is identified by the constant NBGL.

[Note to RFC Editor: please replace NBGL with the IANA provided RTCP XR block type for this block.]

Interval Metric flag (I): 2 bits

This field is used to indicate whether the Burst/Gap Loss metrics are Sampled, Interval or Cumulative metrics:

I=10: Interval Duration - the reported value applies to the most recent measurement interval duration between successive metrics reports.

I=11: Cumulative Duration - the reported value applies to the accumulation period characteristic of cumulative measurements.

I=01: Sampled Value - the reported value is a sampled instantaneous value.

In this document, Burst/Gap Loss Metrics can only be measured over definite intervals, and cannot be sampled. Accordingly, the value I=01, indicating a sampled value, MUST NOT be used. In addition, the value I=00 is reserved and also MUST NOT be used. The block MUST be discarded if the value I=01 or I=00 is received.

Loss and Discard Combination flag (C): 1 bit

The 'C' flag is used to indicate whether combining loss/discard report is needed. This field MUST be set to '1' if the burst gap loss report is present in conjunction with the burst gap discard report in the same compound RTCP packet and MUST be set to '0' otherwise. If the burst gap discard is not sent with the burst gap loss, then the receiver MUST discard the burst gap loss with 'C' flag set to 1. If the 'C' flag is set to 0, then receiver MUST NOT discard the burst gap loss Metrics Block when the burst gap discard is not received.

Reserved (resv): 5 bits

These bits are reserved. They MUST be set to zero by senders and ignored by receivers (See [RFC6709 section 4.2](#)). The block MUST be discarded if the block length is set to a different value.

block length: 16 bits

The length of this report block in 32-bit words, minus one. For the Burst/Gap Loss block, the block length is equal to 5. The block MUST be discarded if the block length is set to a different value.

SSRC of source: 32 bits

As defined in [Section 4.1 of \[RFC3611\]](#).

Threshold: 8 bits

The Threshold is equivalent to Gmin in [[RFC3611](#)], i.e. the number of successive packets that must be received prior to and following a lost packet in order for this lost packet to be regarded as part of a gap.

Sum of Burst Durations (ms): 24 bits

The total duration of bursts of lost packets in the period of the report (Interval or Cumulative).

If the measured value exceeds 0xFFFFFD, the value 0xFFFFFE MUST be reported to indicate an over-range measurement. If the measurement is unavailable, the value 0xFFFFF MUST be reported.

Packets lost in bursts: 24 bits

The total number of packets lost during loss bursts.

If the measured value exceeds 0xFFFFFD, the value 0xFFFFFE MUST be reported to indicate an over-range measurement. If the measurement is unavailable, the value 0xFFFFF MUST be reported.

Total packets expected in bursts: 24 bits

The total number of packets expected during loss bursts (that is, the sum of received packets and lost packets).

If the measured value exceeds 0xFFFFFD, the value 0xFFFFFE MUST be reported to indicate an over-range measurement. If the measurement is unavailable, the value 0xFFFFF MUST be reported.

Number of bursts: 16 bits

The number of bursts in the period of the report (Interval or Cumulative).

If the measured value exceeds 0xFFFFD, the value 0xFFFFE MUST be

reported to indicate an over-range measurement. If the measurement is unavailable, the value 0xFFFF MUST be reported.

Sum of Squares of Burst Durations (ms-squared): 36 bits

The sum of the squares of burst durations (where individual burst durations are expressed in ms) over in the period of the report (Interval or Cumulative). The units for this quantity are milliseconds-squared.

If the measured value exceeds 0xFFFFFFFFD, the value 0xFFFFFFFFE MUST be reported to indicate an over-range measurement. If the measurement is unavailable, the value 0xFFFFFFFF MUST be reported.

3.3. Derived metrics based on reported metrics

The metrics described here are intended to be used as described in this section, in conjunction with information from the Measurement Information block [[RFC6776](#)] (which MUST be present in the same RTCP packet as the Burst/Gap Loss block) and also with the metric "cumulative number of packets lost" provided in standard RTCP [[RFC3550](#)].

These metrics provide information relevant to statistical parameters, including:

- o The fraction of packets lost during bursts (i.e., Burst Loss Rate in [[SUMSTAT](#)]), which can be calculated using the metric "Packets Loss in Bursts" and the metric "Total Packets expected in Bursts" provided in the Burst/Gap Loss metrics block.
- o The fraction of packets lost during gaps (i.e., Gap Loss Rate in [[SUMSTAT](#)]), which can be calculated using the metric "Packets Loss in Bursts" and the metric "Total Packets expected in Bursts" provided in the Burst/Gap Loss metrics block.
- o burst duration mean [[SUMSTAT](#)], which can be calculated using the metric "Packets Loss in Bursts" and the metric "Total Packets expected in Bursts" provided in the Burst/Gap Loss metrics block.
- o burst duration variance [[SUMSTAT](#)], which can be calculated using the metric "Packets Loss in Bursts" and the metric "Total Packets expected in Bursts" provided in the Burst/Gap Loss metrics block.

The details on calculation these parameters in the metrics are described in [[SUMSTAT](#)].

4. Considerations for Voice-over-IP applications

This Metrics Block is applicable to a broad range of RTP applications. Where the metric is used with a Voice-overIP (VoIP) application and the stream repair means is not available, the following considerations apply.

RTCP XR views a call as being divided into bursts, which are periods during which the loss rate is high enough to cause noticeable call quality degradation (generally over 5 percent loss rate), and gaps, which are periods during which lost packets are infrequent and hence call quality is generally acceptable.

If Voice Activity Detection is used the Burst and Gap Duration shall be determined as if silence packets had been sent, i.e. a period of silence in excess of Gmin packets MUST terminate a burst condition.

The recommended value for the threshold Gmin in [[RFC3611](#)] results in a Burst being a period of time during which the call quality is degraded to a similar extent to a typical Pulse-Code Modulation(PCM) Severely Errored Second.

5. SDP Signaling

[RFC3611] defines the use of SDP (Session Description Protocol) [RFC4566] for signaling the use of XR blocks. XR blocks MAY be used without prior signaling.

5.1. SDP rtcp-xr-attrib Attribute Extension

This section augments the SDP [RFC4566] attribute "rtcp-xr" defined in [RFC3611] by providing an additional value of "xr-format" to signal the use of the report block defined in this document.

xr-format =/ xr-bgl-block

xr-bgl-block = "brst-gap-loss"

5.2. Offer/Answer Usage

When SDP is used in offer-answer context, the SDP Offer/Answer usage defined in [RFC3611] for unilateral "rtcp-xr" attribute parameters applies. For detailed usage in Offer/Answer for unilateral parameter, refer to [section 5.2 of \[RFC3611\]](#).

6. IANA Considerations

New block types for RTCP XR are subject to IANA registration. For general guidelines on IANA considerations for RTCP XR, refer to [\[RFC3611\]](#).

6.1. New RTCP XR Block Type value

This document assigns the block type value NBGL in the IANA " RTP Control Protocol Extended Reports (RTCP XR) Block Type Registry " to the "Burst/Gap Loss Metrics Block".

[Note to RFC Editor: please replace NBGL with the IANA provided RTCP XR block type for this block.]

6.2. New RTCP XR SDP Parameter

This document also registers a new parameter "brst-gap-loss" in the " RTP Control Protocol Extended Reports (RTCP XR) Session Description Protocol (SDP) Parameters Registry".

6.3. Contact information for registrations

The contact information for the registrations is:

Qin Wu (sunseawq@huawei.com)

101 Software Avenue, Yuhua District
Nanjing, Jiangsu 210012
China

7. Security Considerations

It is believed that this proposed RTCP XR report block introduces no new security considerations beyond those described in [[RFC3611](#)]. This block does not provide per-packet statistics so the risk to confidentiality documented in [Section 7](#), paragraph 3 of [[RFC3611](#)] does not apply.

8. Contributors

Geoff Hunt wrote the initial draft of this document.

9. Acknowledgments

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Appendix A. Change Log

Note to the RFC-Editor: please remove this section prior to publication as an RFC.

A.1. [draft-ietf-xrblock-rtcp-xr-burst-gap-loss-07](#)

The following are the major changes compared to previous version:

- o Editorial changes based on comments in the WGLC.

A.2. [draft-ietf-xrblock-rtcp-xr-burst-gap-loss-06](#)

The following are the major changes compared to previous version:

- o SDP update based on SDP Directorate Review.
- o Add some texts to get consistent with [RFC6798](#) and Delay draft.

A.3. [draft-ietf-xrblock-rtcp-xr-burst-gap-loss-04](#)

The following are the major changes compared to previous version:

- o Outdated reference update.
- o Editorial changes based on comments that applied to PDV and Delay drafts.

Authors' Addresses

Alan Clark
Telchemy Incorporated
2905 Premiere Parkway, Suite 280
Duluth, GA 30097
USA

Email: alan.d.clark@telchemy.com

Sunshine Zhang (editor)
Shanghai Research Institute of China Telecom Corporation Limited
No.1835, South Pudong Road
Shanghai 200122
China

Email: zhangyx@sttri.com.cn

Jing Zhao
Shanghai Research Institute of China Telecom Corporation Limited
No.1835, South Pudong Road
Shanghai 200122
China

Email: zhaojing@sttri.com.cn

Qin Wu
Huawei
101 Software Avenue, Yuhua District
Nanjing, Jiangsu 210012
China

Email: sunseawq@huawei.com

