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**RTCP XR - IP Video Metrics Report Blocks**  
**draft-ietf-avt-rtcpxr-video-00**

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Abstract

This document defines extensions to the RTCP XR extended report packet type blocks to support the monitoring of video over IP and the associated audio streams, if present, for IPTV and video conferencing endpoint reporting.

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## [1.](#) Introduction

This draft defines several new block types to augment those defined in [RFC3611](#) for use in Quality of Service reporting for video over IP. The new block types defined in this draft are the IP Video Metrics Report Block, and the IP Video Metrics Configuration Block. It is intended to support both the identification of problems affecting performance and the collection of data that may be useful in optimizing system configuration.

Video performance may be measured using zero (no) reference, partial (reduced) reference or full reference. The primary application of this draft is to support the reporting of real-time, in-service performance obtained using a zero or partial reference model however this approach could also be used to support the remote reporting of metrics from a full reference test.

## [2.](#) Definitions

### 2.1 Reporting Endpoint

A report block produced per this draft is produced by the receiving endpoint of an RTP stream, and relates to the quality of the received stream and impairments that may affect perceived quality. A single report block relates to an individual video stream.

## 2.2 Protocol layering

Packet video may be encapsulated in RTP, MPEG-2 Transport or other video transport protocols. Some implementations encapsulate one transport protocol within another, for example MPEG-2 Transport over RTP.

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Video transport protocols may be carried over TCP, UDP, Reliable UDP and may be unicast, multicast or broadcast. Some implementations use combinations of these, for example multicast transmission with unicast retransmission. Forward Error Correction (FEC) may also be used to correct (replace) lost packets.

The video stream comprises a series of I frames, which are intra-frame encoded, and potentially P and B frames, which are interframe encoded. The effects of packet loss can vary considerably depending on the type of frame being impacted.

This draft supports the reporting of metrics related to each layer.

### 2.2 Cumulative and Interval Metrics

Cumulative metrics relate to the entire duration of the session to the point at which metrics are determined and reported, and are typically used to report session quality. Cumulative metrics generally result in a lower volume of data that may need to be stored, as each report supersedes earlier reports.

Interval metrics relate to the period since the last Interval report. Interval data may be easier to correlate with specific network events for which timing is known, and may also be used as a basis for threshold crossing alerts.

Note that interval metrics for the start and end of sessions may be unreliable due to factors such as irregular interval length and the difficulty in knowing when packet transmission started and ended.

### 2.3 Metrics related to packet loss distribution

The distribution of lost packets can have a material impact on the quality of a decoded video stream as packets tend to be lost in high loss periods or bursts.

The terms Burst and Gap are used in a manner consistent with that of RTCP XR ([RFC3611](#)). A Gap is a period of time between Bursts such that any lost or discarded packets or frames are separated by some number of "good" packets or frames. A Burst is a period of time that

fails the test for a Gap, and hence corresponds to a degraded quality period. The recommended value for Gmin in [RFC3611](#) resulted in a Burst being a period of time during which the packet loss/discard rate exceeded 5%. As video is generally more sensitive to packet loss this report block recommends a larger value for Gmin.

Some video decoders do not properly handle out-of-sequence packets and may discard them. The term "discarded" is used to relate to packets that have been discarded due to late arrival or arrival out-of-sequence.

Burst metrics may be used to identify "worst case" settings for FEC or peak bandwidth for retransmission based protocols.

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#### (i) FEC configuration

The burst loss rate represents the average packet loss rate during worst case conditions. If the loss rate correctable by FEC is greater than the burst loss rate, then most bursts of packet loss should be corrected.

#### (ii) Peak bandwidth

Each lost packet that occurs during a burst period would potentially be retransmitted. The peak retransmitted packet rate will therefore be equivalent to the burst packet loss rate.

The term Loss Period is used in the sense defined by IPPM in [RFC3357](#) and relates to a period of consecutive loss.

### 2.4 Absolute and Relative MOS scores

The term MOS (Mean Opinion Score) is used in subjective testing and hence is a range that has a known relationship to "quality". The term can however be confusing when used with services that are not similar. For example, should the MOS score associated with a high definition TV service be the same as that associated with video displayed on a mobile handset? This can make it hard to understand what a MOS score such as 3.1 for a mobile service means - is this the result of degradation or just the result of the smaller display size?

The term Absolute MOS is used in this draft to indicate a MOS score that considers image resolution, frame rate, codec and compression level, the effects of transmission impairments and frame loss concealment, but not the physical size of the display.

The term Relative MOS is used in this draft to indicate a MOS score that is expressed relative to the ideal for this codec and image

For example, a mobile handset service has an Absolute MOS of 3.1 and a Relative MOS of 4.4. This indicates that the service is close to ideal for the application but that some degradation is occurring.

This report block makes use of binary fractions. The terminology used is

Hence 8:8 represents an unsigned number in the range 0.0039 to 255.996.

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### 3.1 Block Description

Header sub-block (Required)

### IP Layer Loss Metrics sub-block (Required)

0								1								2								3							
0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7
Pre-EC Loss Rate								Post-EC Loss Rate																							
Number of IP packets expected																															

RTP Metrics sub-block (Optional)

0								1								2								3							
0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7
SSRC of source																															
Average Network PDV																Peak smoothing PDV															
Loss Rate																Discard Rate															

MPEG Transport Metrics sub-block (Optional)

0								1								2								3							
0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7
Video Stream PID																Audio Stream PID															
Video Stream Loss Rate																Audio Stream Loss Rate															
PCR Jitter																Discard Rate															

Packet Loss/Discard Distribution Metrics Sub-block (Required)

0								1								2								3							
0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7
Burst Duration (ms)																Gmin															
Gap Duration (ms)																															
Burst Loss/Disc Proportion																Gap Loss/Disc Proportion															
Maximum Loss Period																Mean Loss Period															

Video/Audio Metrics sub-block (Required)

0								1								2								3							
0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7
Full Frame Loss Rate																Interpolated Frame Loss Rate															
VSTQ - Transmission Quality																VSCQ - Control Quality															

```

+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|      MOS-A - Audio Quality      |      Reserved      |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|      Absolute MOS-V      |      Relative MOS-V      |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|      Absolute MOS-AV      |      Relative MOS-AV      |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|                               | Video bit rate (bits/sec) |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|                               | Audio bit rate (bits/sec) |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|      A-V Delay (Network I/F)  |      A-V Delay (Video I/F)  |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|      Round Trip Delay (media) |      Round Trip Delay (control) |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+

```

Playout Buffer Metrics sub-block (Required)

```

0           1           2           3
0 1 2 3 4 5 6 7 0 1 2 3 4 5 6 7 0 1 2 3 4 5 6 7 0 1 2 3 4 5 6 7
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
| Playout Interrupt Count      | Mean Playout Interrupt Size  |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|      Playout buffer size      |      Mean buffer level      |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+

```

## 3.2 Header

Implementations **MUST** send the Header block within each RTCP XR Video Metrics report.

### 3.2.1 Block type

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Three Video Performance Reporting Metrics blocks are defined

```

mmm    = Video Metrics- Cumulative
mmm+1  = Video Metrics- Interval
mmm+2  = Video Metrics- Alert

```

The time interval associated with these report blocks is left to the implementation. Spacing of RTCP reports should be in accordance with [RFC3550](#) however the specific timing of RTCP XR Video reports may be determined in response to an internally derived alert such as a threshold crossing.

### 3.2.2 Map field

A Map field indicates the optional sub-blocks present in this report. A 1 indicates that the sub-block is present, and a 0 that the block is absent. If present, the sub-blocks must be in the

sequence defined in this document.  
The bits have the following definitions:

- 0 RTP Metrics block
- 1 MPEG Transport Metrics block
- 2-7 Reserved, set to 0

### 3.2.3 Block Length

The block length indicates the length of this report in 32 bit words and includes the header and any extension octets.

### 3.2.5 Correlation tag

The correlation tag facilitates the correlation of this report block with other call or session related data or endpoint data.

## **3.3 IP Layer loss metrics sub-block**

The IP Layer loss metrics sub-block MUST be present.

This block provides information on IP packet loss, both before and after the effects of error correction.

### 3.3.1 Pre-EC Loss Rate

The proportion of IP packets lost before the effects of error correction (FEC or retransmission), expressed as a binary fraction in 0:16 format.

### 3.3.2 Post-EC Loss Rate

The effective proportion of IP packets after before the effects of error correction, expressed as a binary fraction in 0:16 format.

### 3.3.3 Number of IP Packets Expected

The number of IP packets that the receiving system estimates that it should have received.

## **3.4 RTP Metrics sub-block**

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If RTP is used for media transport, the RTP Metrics sub-block MUST be present and if present MUST be indicated in the Map field.

This block provides information on the effects of IP transmission impairments on the RTP stream.

### 3.4.1 Source SSRC

The SSRC associated with the RTP stream to which this report block relates.



#### 3.4.2 Average network PDV

The average delay variation of RTP packets due to the effects of network congestion and buffering.

#### 3.4.3 Peak smoothing PDV

The peak delay variation due to smoothing of the video packet transmission rate, either by the sending system or network based rate control. This should be determined by comparing the variation in arrival time to the variation in RTP time stamp, and observing any periodicity in the resulting sequence of delay variations.

#### 3.4.4 Loss Rate

The proportion of RTP packets lost in the network, after the effects of any error correction or retransmission. This should be determined by comparing the variation in in RTP time stamp, and removing any periodicity in the resulting sequence of delay variations.

#### 3.4.5 Discard Rate

The proportion of RTP packets discarded due to out-of-sequence, late or early arrival.

### **3.5 MPEG-2 Transport Metrics sub-block**

The MPEG-2 Transport Metrics sub-block MUST be present if MPEG Transport is used, and if present MUST be indicated in the Map field.

This block contains a number of metrics associated with MPEG transport stream packets.

#### 3.5.1 Video Stream Program ID

The Program ID (PID) associated with the video stream.

#### 3.5.2 Audio Stream Program ID

The Program ID (PID) associated with the audio stream

#### 3.5.3 Loss Rate

The proportion of MPEG Transport Stream packets lost in the network, after the effects of any error correction or retransmission.

#### 3.5.4 Discard Rate

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The proportion of MPEG Transport Stream packets discarded due to late or early arrival.

#### 3.5.5 PCR Jitter

The average PCR (Program Clock Reference) jitter level in milliseconds for this MPEG Transport Stream.

### **3.6 Packet Loss/Discard Distribution Metrics sub-block**

The Packet Loss/Discard Distribution Metrics sub-block MUST be present.

This block contains metrics that describe the time distribution of lost and discarded packets after the effects of any error correction.

#### **3.6.1 Burst duration**

The duration of bursts of lost and discarded RTP packets expressed in milliseconds.

#### **3.6.2 Gmin Threshold**

The Gmin threshold associated with the definition of bursts and gaps.

#### **3.6.3 Gap duration**

The mean duration of gaps between bursts expressed in milliseconds.

#### **3.6.4 Burst loss/discard proportion**

The proportion of frames lost or discarded during burst periods expressed as a binary fraction.

#### **3.6.5 Gap loss/discard proportion**

The proportion of frames lost or discarded during gap periods expressed as a binary fraction.

#### **3.6.6 Maximum Loss Period**

The maximum number of consecutive lost packets during this session.

#### **3.6.7 Mean Loss Period**

The mean number of consecutive lost packets during this session.

### **3.7 Video/Audio Metrics sub-block**

The Video/Audio Metrics sub-block MUST be present.

The metrics in this block provide information on the quality of the video stream.

#### **3.7.1 Full frame packet loss rate**

The packet loss rate that affected full or intra-frame encoded video frames (I frames).

#### **3.7.2 Interpolated frame packet loss rate**

The packet loss rate that affected interpolated or inter-frame encoded video frames (B/P frames).

### 3.7.3 VSTQ - Video Service Transmission Quality

The video service transmission quality expressed as a score in the range 0.0 to 50.0. This is a codec independent measure of the ability of the bearer channel to support reliable video.

### 3.7.4 VSCQ - Video Service Control Plane Quality

The video service control plane (trick play) quality expressed as a score in the range 0.0 to 50.0. This is a measure that is related to the performance of the video stream control channel.

### 3.7.5 MOS-A Audio Quality

The video service audio quality expressed as a score in the range 1.0 to 5.0. This is an audio codec dependant measure that is related to the subjective quality of the decoded audio stream(s). (ATIS)

### 3.7.6 Absolute MOS-V Picture Quality

The absolute picture quality expressed as a score in the range 1.0 to 5.0. This is a codec dependant measure that is related to the subjective quality of the decoded video stream and considers the effects of codec, loss, bit rate/ quantization level, image resolution and frame loss concealment.

### 3.7.7 Relative MOS-V Picture Quality

Picture quality expressed as a score relative to an ideal picture with the same configuration.

### 3.7.8 Absolute MOS-AV Multimedia Quality

The multimedia quality expressed as a score in the range 1.0 to 5.0. This is a composite audio/video measure that is related to the overall subjective user experience and considers picture quality, audio quality and audio/video synchronization.

### 3.7.9 Relative MOS-AV Multimedia Quality

Multimedia quality expressed as a score relative to an ideal video and audio with the same configuration.

### 3.7.10 Video bit rate

The short term average bit rate of the video codec.

### 3.7.11 Audio bit rate

The short term average bit rate of the audio codec.

### 3.7.12 Audio-Video Delay (Network Interface)

The relative delay between audio and video measured before the decoder and expressed in milliseconds

### 3.7.13 Audio-Video Delay (Video Interface)

The relative delay between audio and video measured after the decoder and expressed in milliseconds

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### 3.7.14 Round Trip Delay (Media)

The round trip delay for the media path, required only for interactive video sessions.

### 3.7.15 Round Trip Delay (Control)

The round trip delay for the video control (trick play) path.

## **3.8 Playout Buffer Metrics sub-block**

**The Playout Buffer Metrics sub-block MUST be present.**

### 3.8.1 Playout Interruption Count

The number of interruptions in video playout that have occurred due to playout buffer starvation or excessive packet loss.

### 3.8.2 Mean Playout Interruption Size

The mean size of interruptions in playout, expressed in multiples of 100 milliseconds

### 3.8.3 Playout Buffer Size

The playout buffer size, expressed in multiples of 100 milliseconds

### 3.8.4 Mean Buffer Size

The average playout buffer size expressed in multiples of 100 milliseconds

## 4. RTCP XR Video Metrics - Compact Report Block

### 4.1 Block description

This block provides a compact alternative to the Video Metrics report block for bandwidth or MTU size constrained applications.

0								1								2								3							
0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7	0	1	2	3	4	5	6	7
BT=N								reserved								block length=9															
Stream ID																															
Pre-EC Loss Rate																Post-EC Loss Rate															
Discard Rate																Burst density								Gap density							
Burst duration																Gap duration															
Round trip delay																Peak Smoothing Jitter															
Network Jitter																Relative MOS-V								Abs MOS-V							
Relative MOS-AV								Abs MOS-AV								MOS-A								A-V Delay							
Playout Interrupt Count																Mean Playout Interrupt Size															
Playout buffer size																Mean buffer level															

### 4.2 Header

Three Compact Video Report Blocks are defined

mmm+3 = Compact Video Metrics- Cumulative  
 mmm+4 = Compact Video Metrics- Interval

mmm+5 = Compact Video Metrics- Alert

The time interval associated with these report blocks is left to the implementation. Spacing of RTCP reports should be in accordance with [RFC3550](#) however the specific timing of RTCP XR Video reports may be determined in response to an internally derived alert such as a threshold crossing.

### [4.3](#) Metrics

#### [4.3.1](#) Pre-EC Loss Rate

Pre-Error Correction Loss Rate is defined in [Section 3.3.1](#).

#### [4.3.2](#) Post-EC Loss Rate

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Post-Error Correction Loss Rate is defined in [Section 3.3.2](#).

#### [4.3.3](#) Discard Rate

Discard Rate is defined in [Section 3.4.5](#).

#### [4.3.4](#) Burst Density

Burst Density is defined in [Sections 3.6.4](#)

#### [4.3.5](#) Gap Density

Gap Density is defined in [Section 3.6.5](#).

#### [4.3.6](#) Burst Duration

Burst Duration is defined in [Section 3.6.1](#)

#### [4.3.7](#) Gap Duration

Gap Duration is defined in [Section 3.6.3](#)

#### [4.3.8](#) Round Trip Delay

Round Trip Delay is defined in [Section 3.7.14](#).

#### [4.3.9](#) Peak Smoothing jitter

Smoothing jitter is defined in [section 3.4.3](#)

#### [4.3.10](#) Average Network Jitter

Average network Jitter is defined in [Section 3.4.2](#).

#### [4.3.11](#) Relative MOS-V

MOS-V is defined in [Section 3.7.7](#).

#### [4.3.12](#) Absolute MOS-V

MOS-V is defined in [Section 3.7.6](#).

#### [4.3.13](#) Relative MOS-AV

MOS-AV is defined in [Section 3.7.9](#).

#### [4.3.14](#) Absolute MOS-AV

MOS-AV is defined in [Section 3.7.8](#).

#### [4.3.15](#) MOS-A

MOS-A is defined in [Section 3.7.5](#).

#### [4.3.16](#) A-V Delay

Audio-Video Delay is defined in [Section 3.7.12](#).

#### [4.3.16](#) Playout Interrupt count

Playout Interrupt Count is defined in [Section 3.8.1](#).

#### [4.3.15](#) Playout Interrupt size

Playout Interrupt Size is defined in [Section 3.8.2](#).

#### [4.3.14](#) Playout buffer size

Playout buffer size is defined in [Section 3.8.3](#).

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#### [4.3.17](#) Mean buffer size

Mean playout buffer size is defined in [Section 3.8.4](#).

### [5](#). RTCP XR Video Metrics Configuration Block

This block type provides a flexible means to describe the algorithms used for video quality calculation and other data. This block need only be exchanged occasionally, for example sent once at the start of a session.

Header sub-block

```

0                               1                               2                               3
0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|      BT=N      |      Map      |      block length      |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|                                     SSRC of source         |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
```

Correlation tag

```

0                               1                               2                               3
0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|                                     SSRC of source         |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
```

```

Algorithm sub-block
  0                               1                               2                               3
  0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
| Alg type      | Descriptor len|      Algorithm descriptor... |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|                ... Algorithm descriptor                |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+

```

5.1 Header

Implementations MUST send the Header block within each Video Metrics Configuration report.

5.1.1 Block type  
One Video Metrics Configuration block is defined

mmm+6 = Video Metric Configuration Block

The time interval associated with these report blocks is left to the implementation. Spacing of RTCP reports should be in accordance with [RFC3550](#) however the specific timing of RTCP XR Video reports may

be determined in response to an internally derived alert such as a threshold crossing.

5.1.2 Map field  
A Map field indicates the optional sub-blocks present in this report. A 1 indicates that the sub-block is present, and a 0 that the block is absent. If present, the sub-blocks must be in the sequence defined in this document.  
The bits have the following definitions:

- 0 Correlation Tag
- 1 Algorithm Descriptor 1
- 2 Algorithm Descriptor 2
- 3 Algorithm Descriptor 3
- 4 Algorithm Descriptor 4
- 5 Vendor Specific Extension
- 6-7 Reserved, set to 0

5.1.3 Block Length  
The block length indicates the length of this report in 32 bit words and includes the header and any extension octets.

5.1.4 SSRC



The SSRC of the stream to which this report relates.

## **5.2 Correlation Tag**

The Correlation Tag sub-block MAY be present and if present MUST be indicated in the map field. This tag facilitates the correlation of the RTCP XR Video Metrics report blocks with other session-related data, session-related data or endpoint data.

An example use case is for an endpoint may convey its version of a session identifier or a global session identifier via this tag. A flow measurement tool (sniffer) that is not session-aware can then forward the RTCP XR reports along with this correlation tag to network management. Network management can then use this tag to correlate this report with other diagnostic information such as session detail records.

The Tag Type indicates the use of the correlation tag. The following values are defined:

- 0: IMS Charging Identity (ICID) subfield of the P-Charging-Vector header specified in [RFC 3455](#).
- 1: Globally unique ID as specified in ITU-T H.225.0 (Table 20/H.225.0).
- 2: Conference Identifier, per ITU-T H.225.0 (Table 20/H.225.0).

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- 3: SIP Call-ID as defined in [RFC 3261](#).
- 4: PacketCable Billing Call ID (BCID).
- 5: Text string using the US-ASCII character set.
- 6: Octet sting.
- 7-255: Future growth.

Although the intent of this RFC is to list all currently known values of usable correlation tags, it is possible that new values may be defined in the future. An IANA registry of correlation tags is recommended.

The tag length indicates the overall length of the sub-block in 32 bit words and includes the tag type and length fields.

### 5.3 Algorithm description

The Algorithm Description sub-block MAY be present however if present MUST be indicated in the MAP field

The Algorithm descriptor is a bit field which indicates which algorithm is being described. The bits are defined as:-

Bit 0:	MOS-LQ Algorithm
Bit 1:	MOS-CQ Algorithm
Bit 2:	R-LQ Algorithm
Bit 3:	R-CQ Algorithm
Bit 4:	Video Monitoring Algorithm
Bit 5:	Audio Monitoring Algorithm
Bit 6:	Multimedia Monitoring Algorithm
Bit 7:	Transmission Quality Monitoring Algorithm

The descriptor length gives the overall length of the descriptor in 32 bit words and includes the algorithm descriptor and length fields.

The algorithm descriptor is a text field that contains the description or name of the algorithm. If the algorithm name is shorter than the length of the field then the trailing octets must be set to 0x00.

For example, an implementation may report:

Algorithm descriptor = 0x10	- Video estimation algorithm
Descriptor length = 3	- 3 words
Descriptor = "Alg X" 0x00	- description

## 6. Summary

This draft defines a full and a compact RTCP XR report block for video quality reporting. This is intended for in-service monitoring of video streaming, IPTV and IP videoconferencing services to provide

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real time performance feedback and support performance management.

## 7. IANA Considerations

The block type "mmm" will need to be replaced with an IANA assigned number within those allocated for RTCP XR report blocks ([RFC 3611](#)).

## 8. Security Considerations

RTCP reports can contain sensitive information since they can provide

information about the nature and duration of a session established between two endpoints. As a result, any third party wishing to obtain this information should be properly authenticated and the information transferred securely.

## **9. Acknowledgments**

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## **10. Informative References**

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