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# RTP Stream Identifier Source Description (SDES) draft-ietf-avtext-rid-03

#### Abstract

This document defines and registers two new RTCP SDES items. One, named RtpStreamId, is used for unique identification of RTP streams. The other, RepairedRtpStreamId, can be used to identify which stream a redundancy RTP stream is to be used to repair.

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### 1. Introduction

RTP sessions frequently consist of multiple streams, each of which is identified at any given time by its SSRC; however, the SSRC associated with a stream is not guaranteed to be stable over its lifetime. Within a session, these streams can be tagged with a number of identifiers, including CNAMEs and MSIDs [I-D.ietf-mmusic-msid]. Unfortunately, none of these have the proper ordinality to refer to an individual stream; all such identifiers can appear in more than one stream at a time. While approaches that use unique Payload Types (PTs) per stream have been used in some applications, this is a semantic overloading of that field, and one for which its size is inadequate: in moderately complex systems that use PT to uniquely identify every potential combination of codec configuration and unique stream, it is possible to simply run out of values.

To address this situation, we define a new RTCP SDES identifier, RtpStreamId, that uniquely identifies a single RTP stream. A key motivator for defining this identifier is the ability to differentiate among different encodings of a single Source Stream that are sent simultaneously (i.e., simulcast). This need for unique identification extends to dependent streams (i.e., layers used by a layered codec).

At the same time, when redundancy RTP streams are in use, we also need an identifier that connects such streams to the RTP stream for which they are providing redundancy. For this purpose, we define an additional SDES identifier, RepairedRtpStreamId. This identifier can appear only in packets associated with a redundancy RTP stream. They carry the same value as the RtpStreamId of the RTP stream that the redundant RTP stream is correcting.

# 2. Terminology

In this document, the terms "source stream", "encoded stream," "RTP stream", "source RTP stream", "dependent stream", "received RTP stream", and "redundancy RTP stream" are used as defined in [RFC7656].

## 3. Usage of RtpStreamId and RepairedRtpStreamId in RTP and RTCP

The RTP fixed header includes the payload type number and the SSRC values of the RTP stream. RTP defines how you de-multiplex streams within an RTP session; however, in some use cases, applications need further identifiers in order to effectively map the individual RTP Streams to their equivalent payload configurations in the SDP.

This specification defines two new RTCP SDES items [RFC3550]. The first item is 'RtpStreamId', which is used to carry RTP stream identifiers within RTCP SDES packets. This makes it possible for a receiver to associate received RTP packets (identifying the RTP stream) with a media description having the format constraint specified. The second is 'RepairedRtpStreamId', which can be used in redundancy RTP streams to indicate the RTP stream repaired by a redundancy RTP stream.

To be clear: the value carried in a RepairedRtpStreamId will always match the RtpStreamId value from another RTP stream in the same session. For example, if a source RTP stream is identified by RtpStreamId "A", then any redundancy RTP stream that repairs that source RTP stream will contain a RepairedRtpStreamId of "A" (if this mechanism is being used to perform such correlation). These redundant RTP streams may also contain their own unique RtpStreamId.

This specification also uses the RTP header extension for RTCP SDES items [I-D.ietf-avtext-sdes-hdr-ext] to allow carrying RtpStreamId and RepairedRtpStreamId values in RTP packets. This allows correlation at stream startup, or after stream changes where the use of RTCP may not be sufficiently responsive.

# 3.1. RTCP 'RtpStreamId' SDES Extension

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
+-	+-	+	+	+ - +	<b>⊢</b> – ⊣	<b>⊦</b> – ⊣	<b>⊢</b> – +		<del>-</del>	<b>-</b> - +	<b>-</b> - +	+		<del>-</del>	<b>-</b> - +	<b>⊦</b> – ⊣	<b>⊦</b> – ⊣	<b>-</b> - +	<b>-</b> -		<b>⊢</b> – +	+	<b>-</b> - +	<b>⊦</b> – ⊣	<b>-</b> - +	<b>⊢</b> – ⊣	1	+	- +	+	· - +
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+-	+-	+	+	<del> </del>	⊢ <b>–</b> ⊣	<b>⊢</b> – ⊣	<b>⊢</b> – +	<del>-</del>	H - H	H - H	H - H	1	H - H	H - H	H - H	<b>⊢</b> – ⊣	<b>⊢</b> – ⊣	H - H	H - H	H - H	H – H	1	H - H	<b>⊢</b> – ⊣	H - H	<b>⊢</b> – ⊣	1	+	+	+	· - +

The RtpStreamId payload is UTF-8 encoded and is not null-terminated.

RFC EDITOR NOTE: Please replace TBD with the assigned SDES identifier value.

# 3.2. RTCP 'RepairedRtpStreamId' SDES Extension

6	1	L 2	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	
+-	+-	+	-+	- +			<b>⊢</b> – +	<b>⊦</b> – ⊣	<b>⊦</b> – ⊣	<del>-</del>	<b>-</b> - +	<b>⊦</b> – ⊣	+		<b>-</b> -		<b>-</b> -	<b>-</b> - +	<b>+</b> - +	<b>-</b> -	<del> </del>	<b>⊢</b> – +	+	<b>⊦</b> – ⊣	<b>⊦</b> – ⊣	<b>-</b> - +	<b>⊢</b> – ⊣	+		+	+	+	
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The RepairedRtpStreamId payload is UTF-8 encoded and is not null-terminated.

RFC EDITOR NOTE: Please replace TBD with the assigned SDES identifier value.

# 3.3. RTP 'RtpStreamId' and 'RepairedRtpStreamId' Header Extensions

Because recipients of RTP packets will typically need to know which streams they correspond to immediately upon receipt, this specification also defines a means of carrying RtpStreamId and RepairedRtpStreamId identifiers in RTP extension headers, using the technique described in [I-D.ietf-avtext-sdes-hdr-ext].

As described in that document, the header extension element can be encoded using either the one-byte or two-byte header, and the identification-tag payload is UTF-8 encoded, as in SDP.

As the identifier is included in an RTP header extension, there should be some consideration given to the packet expansion caused by the identifier. To avoid Maximum Transmission Unit (MTU) issues for the RTP packets, the header extension's size needs to be taken into account when the encoding media. Note that set of header extensions included in the packet needs to be padded to the next 32-bit boundary [RFC5285].

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In many cases, a one-byte identifier will be sufficient to distinguish streams in a session; implementations are strongly encouraged to use the shortest identifier that fits their purposes.

#### 4. IANA Considerations

### 4.1. New RtpStreamId SDES item

RFC EDITOR NOTE: Please replace RFCXXXX with the RFC number of this document.

RFC EDITOR NOTE: Please replace TBD with the assigned SDES identifier value.

This document adds the RtpStreamId SDES item to the IANA "RTCP SDES item types" registry as follows:

Value: TBD

Abbrev.: RtpStreamId

Name: RTP Stream Identifier

Reference: RFCXXXX

# 4.2. New RepairRtpStreamId SDES item

RFC EDITOR NOTE: Please replace RFCXXXX with the RFC number of this document.

RFC EDITOR NOTE: Please replace TBD with the assigned SDES identifier value.

This document adds the RepairedRtpStreamId SDES item to the IANA "RTCP SDES item types" registry as follows:

Value: TBD

Abbrev.: RepairedRtpStreamId

Name: Repaired RTP Stream Identifier

Reference: RFCXXXX

# **5**. Security Considerations

The actual identifiers used for RtpStreamIds (and therefore RepairedRtpStreamIds) are expected to be opaque. As such, they are not expected to contain information that would be sensitive, were it observed by third-parties.

# Acknowledgements

Many thanks for review and input from Cullen Jennings, Magnus Westerlund, Colin Perkins, Peter Thatcher, Jonathan Lennox, and Paul Kyzivat.

### 7. References

# 7.1. Normative References

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### 7.2. Informative References

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