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A General Mechanism for RTP Header Extensions
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Abstract

This document provides a general mechanism to use the header extension feature of RTP (the Real-Time Transport Protocol). It provides the option to use a small number of small extensions in each RTP packet, where the universe of possible extensions is large and registration is de-centralized. The actual extensions in use in a session are signaled in the setup information for that session. This document obsoletes RFC5285.

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1. Introduction

The RTP specification [RFC3550] provides a capability to extend the RTP header. It defines the header extension format and rules for its use in Section 5.3.1. The existing header extension method permits at most one extension per RTP packet, identified by a 16-bit identifier and a 16-bit length field specifying the length of the header extension in 32-bit words.

This mechanism has two conspicuous drawbacks. First, it permits only one header extension in a single RTP packet. Second, the specification gives no guidance as to how the 16-bit header extension identifiers are allocated to avoid collisions.

This specification removes the first drawback by defining a backward-compatible and extensible means to carry multiple header extension elements in a single RTP packet. It removes the second drawback by defining that these extension elements are named by URIs, defining an IANA registry for extension elements defined in IETF specifications, and a Session Description Protocol (SDP) method for mapping between the naming URIs and the identifier values carried in the RTP packets.

This header extension applies to RTP/AVP (the Audio/Visual Profile) and its extensions.

This document obsoletes [RFC5285] and removes a limitation from RFC5285 that did not allow sending both one-byte and two-byte header extensions in the same RTP stream

2. Requirements Notation

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 [RFC2119].

3. Design Goals

The goal of this design is to provide a simple mechanism whereby multiple identified extensions can be used in RTP packets, without the need for formal registration of those extensions but nonetheless avoiding collision.

This mechanism provides an alternative to the practice of burying associated metadata into the media format bit stream. This has often been done in media data sent over fixed-bandwidth channels. Once this is done, a decoder for the specific media format needs to extract the metadata. Also, depending on the media format, the metadata can be added at the time of encoding the media so that the bit-rate used for the metadata is taken into account. But the metadata can be unknown at that time. Inserting metadata at a later time can cause a decode and re-encode to meet bit-rate requirements.

In some cases, a more appropriate, higher-level mechanism may be available, and if so, it can be used. For cases where a higher-level mechanism is not available, it is better to provide a mechanism at the RTP level than have the metadata be tied to a specific form of media data.

4. Packet Design

4.1. General

The following design is fit into the "header extension" of the RTP extension, as described above.

The presence and format of this header extension and its contents are negotiated or defined out-of-band, such as through signaling (see below for SDP signaling). The 16-bit identifier for the two forms of RTP extension defined here is only an architectural constant (e.g., for use by network analyzers); it is the negotiation/definition (e.g., in SDP) that is the definitive indication that this header extension is present.

The RTP specification [RFC3550] states that RTP "is designed so that the header extension may be ignored by other interoperating implementations that have not been extended". The intent of this restriction is that RTP header extensions MUST NOT be used to extend RTP itself in a manner that is backwards incompatible with non-extended implementations. For example, a header extension is not allowed to change the meaning or interpretation of the standard RTP header fields, or of the RTCP Control Protocol (RTCP). Header extensions MAY carry metadata in addition to the usual RTP header information, provided the RTP layer can function if that metadata is missing. For example, RTP header extensions can be used to carry data that's also sent in RTCP, as an optimisation to lower latency, since they'll fall back to the original, non-optimised, behaviour if the header extension is not present. The use of header extensions to convey information that will, if missing, disrupt the behaviour of a higher layer application that builds on top of RTP is only acceptable if this doesn't affect interoperability at the RTP layer. For example, applications that use the SDP BUNDLE extension with the MID RTP header extension [I-D.ietf-mmusic-sdp-bundle-negotiation] to correlate RTP streams with SDP m= lines likely won't work with full functionality if the MID is missing, but the operation of the RTP layer of those applications will be unaffected. Support for RTP header extensions based on this memo is negotiated using, for example, SDP offer answer [RFC3264]; intermediaries aware of the RTP header extensions are advised to be cautious when removing or generating RTP header extensions see section 4.7 of [RFC7667].

The RTP header extension is formed as a sequence of extension elements, with possible padding. Each extension element has a local identifier and a length. The local identifiers MAY be mapped to a larger namespace in the negotiation (e.g., session signaling).

4.1.1. Transmission Considerations

As is good network practice, data should only be transmitted when needed. The RTP header extension SHOULD only be present in a packet if that packet also contains one or more extension elements, as defined here. An extension element SHOULD only be present in a packet when needed; the signaling setup of extension elements indicates only that those elements can be present in some packets, not that they are in fact present in all (or indeed, any) packets.

Some general considerations for getting the header extensions delivered to the receiver:

1. The probability for packet loss and burst loss determine how many repetitions of the header extensions will be required to reach a targeted delivery probability, and if burst loss is likely, what distribution would be needed to avoid getting all repetitions of the header extensions lost in a single burst.
2. If a set of packets are all needed to enable decoding, there is commonly no reason for including the header extension in all of these packets, as they share fate. Instead, at most one instance of the header extension per independently decodable set of media data would be a more efficient use of the bandwidth.
3. How early the Header Extension item information is needed, from the first received RTP data or only after some set of packets are received, can guide if the header extension(s) should be in all of the first N packets or be included only once per set of packets, for example once per video frame.
4. The use of RTP level robustness mechanisms, such as RTP retransmission [RFC4588], or Forward Error Correction, e.g., [RFC5109] may treat packets differently from a robustness perspective, and header extensions should be added to packets that get a treatment corresponding to the relative importance of receiving the information.

As a summary, the number of header extension transmissions should be tailored to a desired probability of delivery taking the receiver population size into account. For the very basic case, N repetitions of the header extensions should be sufficient, but may not be optimal. N is selected so that the header extension target delivery probability reaches $1 - P^N$, where P is the probability of packet loss. For point to point or small receiver populations, it might also be possible to use feedback, such as RTCP, to determine when the information in the header extensions has reached all receivers and stop further repetitions. Feedback that can be used includes the

RTCP XR Loss RLE report block [RFC3611], which will indicate successful delivery of particular packets. If the RTP/AVPF Transport Layer Feedback Messages for generic NACK [RFC4585] is used, it can indicate the failure to deliver an RTP packet with the header extension, thus indicating the need for further repetitions. The normal RTCP report blocks can also provide an indicator of successful delivery, if no losses are indicated for a reporting interval covering the RTP packets with the header extension. Note that loss of an RTCP packet reporting on an interval where RTP header extension packets were sent, does not necessarily mean that the RTP header extension packets themselves were lost.

4.1.2. Header Extension Type Considerations

Each extension element in a packet has a local identifier (ID) and a length. The local identifiers present in the stream MUST have been negotiated or defined out-of-band. There are no static allocations of local identifiers. Each distinct extension MUST have a unique ID. The ID value 0 is reserved for padding and MUST NOT be used as a local identifier.

An extension element with an ID value equal 0 MUST NOT have len field greater than 0. If such an extension element is encountered, its length field MUST be ignored, processing of the entire extension MUST terminate at that point, and only the extension elements present prior to the element with ID 0 and len field greater than 0 SHOULD be considered.

There are two variants of the extension: one-byte and two-byte headers. Since it is expected that (a) the number of extensions in any given RTP session is small and (b) the extensions themselves are small, the one-byte header form is preferred and MUST be supported by all receivers. A stream MUST contain only one-byte or two-byte headers unless it is known that all recipients support mixing, either by offer/answer negotiation (see section 6) or by out-of-band knowledge. Each RTP packet with an RTP header extension following this specification will indicate if it contains one or two byte header extensions through the use of the "defined by profile" field. Extension element types that do not match the header extension format, i.e. one- or two-byte, MUST NOT be used in that RTP packet. Transmitters SHOULD NOT use the two-byte form when all extensions are small enough for the one-byte header form. Transmitters that intend to send the two-byte form SHOULD negotiate the use of IDs above 14 if they want to let the Receivers know that they intend to use two-byte form, for example if the RTP header extension is longer than 16 bytes. A transmitter may be aware that an intermediary may add RTP header extensions in this case, the transmitter SHOULD use two-byte form.

A sequence of extension elements, possibly with padding, forms the header extension defined in the RTP specification. There are as many extension elements as fit into the length as indicated in the RTP header extension length. Since this length is signaled in full 32-bit words, padding bytes are used to pad to a 32-bit boundary. The entire extension is parsed byte-by-byte to find each extension element (no alignment is needed), and parsing stops at the earlier of the end of the entire header extension, or in one-byte headers only case, on encountering an identifier with the reserved value of 15.

In both forms, padding bytes have the value of 0 (zero). They MAY be placed between extension elements, if desired for alignment, or after the last extension element, if needed for padding. A padding byte does not supply the ID of an element, nor the length field. When a padding byte is found, it is ignored and the parser moves on to interpreting the next byte.

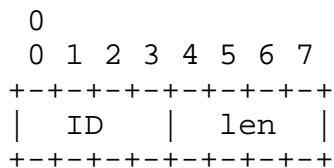
Note carefully that the one-byte header form allows for data lengths between 1 and 16 bytes, by adding 1 to the signaled length value (thus, 0 in the length field indicates 1 byte of data follows). This allows for the important case of 16-byte payloads. This addition is not performed for the two-byte headers, where the length field signals data lengths between 0 and 255 bytes.

Use of RTP header extensions will reduce the efficiency of RTP header compression, since the header extension will be sent uncompressed unless the RTP header compression module is updated to recognize the extension header. If header extensions are present in some packets, but not in others, this can also reduce compression efficiency by requiring an update to the fixed header to be conveyed when header extensions start or stop being sent. The interactions of the RTP header extension and header compression is explored further in [RFC2508] and [RFC3095].

4.2. One-Byte Header

In the one-byte header form of extensions, the 16-bit value required by the RTP specification for a header extension, labeled in the RTP specification as "defined by profile", MUST have the fixed bit pattern 0xBEDE (the first version of this specification was written on the feast day of the Venerable Bede).

Each extension element MUST start with a byte containing an ID and a length:

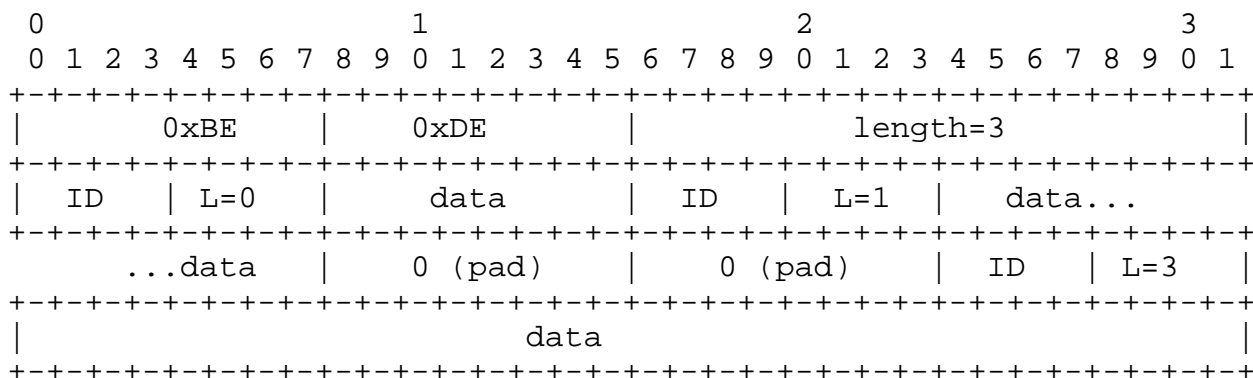


The 4-bit ID is the local identifier of this element in the range 1-14 inclusive. In the signaling section, this is referred to as the valid range.

The local identifier value 15 is reserved for future extension and MUST NOT be used as an identifier. If the ID value 15 is encountered, its length field MUST be ignored, processing of the entire extension MUST terminate at that point, and only the extension elements present prior to the element with ID 15 SHOULD be considered.

The 4-bit length is the number minus one of data bytes of this header extension element following the one-byte header. Therefore, the value zero in this field indicates that one byte of data follows, and a value of 15 (the maximum) indicates element data of 16 bytes. (This permits carriage of 16-byte values, which is a common length of labels and identifiers, while losing the possibility of zero-length values -- which would often be padded anyway.)

An example header extension, with three extension elements, and some padding follows:



4.3. Two-Byte Header

In the two-byte header form, the 16-bit value defined by the RTP specification for a header extension, labeled in the RTP specification as "defined by profile", is defined as shown below.

```

      0                               1
      0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|           0x100           |appbits|
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+

```

The appbits field is 4 bits that are application-dependent and MAY be defined to be any value or meaning, and are outside the scope of this specification. For the purposes of signaling, this field is treated as a special extension value assigned to the local identifier 256. If no extension has been specified through configuration or signaling for this local identifier value 256, the appbits field SHOULD be set to all 0s by the sender and MUST be ignored by the receiver.

Each extension element starts with a byte containing an ID and a byte containing a length:

```

      0                               1
      0 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+---+
|           ID           | length |
+---+---+---+---+---+---+---+---+---+---+---+---+---+---+

```

The 8-bit ID is the local identifier of this element in the range 1-255 inclusive. In the signaling section, the range 1-256 is referred to as the valid range, with the values 1-255 referring to extension elements, and the value 256 referring to the 4-bit field 'appbits' (above). Note that there is one ID space for both one-byte and two-byte form. This means that the lower values (1-14) can be used in the 4-bit ID field in the one-byte header format with the same meanings.

The 8-bit length field is the length of extension data in bytes not including the ID and length fields. The value zero indicates there is no data following.

An extension URI with the same attributes MUST NOT appear more than once applying to the same stream, i.e., at session level or in the declarations for a single stream at media level. (The same extension can, of course, be used for several streams, and can appear with different extensionattributes for the same stream.)

For extensions defined in RFCs, the URI used SHOULD be a URN starting "urn:ietf:params:rtp-hdext:" and followed by a registered, descriptive name.

The registration requirements are detailed in the IANA Considerations section, below.

An example (this is only an example), where 'avt-example-metadata' is the hypothetical name of a header extension, might be:

```
urn:ietf:params:rtp-hdext:avt-example-metadata
```

An example name not from the IETF (this is only an example) might be:

```
http://example.com/082005/ext.htm#example-metadata
```

The mapping MAY be provided per media stream (in the media-level section(s) of SDP, i.e., after an "m=" line) or globally for all streams (i.e., before the first "m=" line, at session level). The definitions MUST be either all session level or all media level; it is not permitted to mix the two styles. In addition, as noted above, the IDs used MUST be unique in each media section of the SDP, or unique in the session for session-level SDP declarations.

Each local identifier potentially used in the stream is mapped to an extension identified by a URI using an attribute of the form:

```
a=extmap:<value>["/"<direction>] <URI> <extensionattributes>
```

where <URI> is a URI, as above, <value> is the local identifier (ID) of this extension and is an integer in the valid range (0 is reserved for padding in both forms, and 15 is reserved in the one-byte header form, as noted above), and <direction> is one of "sendonly", "recvonly", "sendrecv", or "inactive" (without the quotes) with relation to the device being configured.

The formal BNF syntax is presented in a later section of this specification.

Example:

```
a=extmap:1 http://example.com/082005/ext.htm#ttime
```

```
a=extmap:2/sendrecv http://example.com/082005/ext.htm#xmeta short
```

When SDP signaling is used for the RTP session, it is the presence of the 'extmap' attribute(s) that is diagnostic that this style of header extensions is used, not the magic number indicated above.

6. SDP Signaling for support of mixed one byte and two bytes header extensions.

In order to allow for backward interoperability with systems that do not support mixing of one byte and two bytes header extensions this document defines the "a=extmap-allow-mixed" Session Description Protocol (SDP) [RFC4566] attribute to indicate if the participant is capable of supporting this new mode. The attribute takes no value. This attribute can be used at the session or media levels. A participant that proposes the use of this mode SHALL itself support the reception of mixed one byte and two bytes header extensions.

If offer/answer [RFC3264] is supported and used, the negotiation for mixed one byte and two bytes extension MUST be negotiated using offer/answer [RFC3264]. In the absence of negotiations using offer/answer, for example when declarative SDP is used, mixed headers MUST NOT occur unless the transmitter has some (out of band) knowledge that all potential recipients support this mode.

The formal definition of this attribute is:

Name: extmap-allow-mixed

Value:

Usage Level: session, media

Charset Dependent: no

Example:

```
a=extmap-allow-mixed
```

When doing SDP Offer/Answer [RFC3264] an offering client that wishes to use both one and two bytes extensions MUST include the attribute "a= extmap-allow-mixed " in the SDP offer. If "a= extmap-allow-mixed " is present in the offer SDP, the answerer that supports this mode and wishes to use it SHALL include the "a=extmap-allow-mixed " attribute in the answer. In the cases where the attribute has been excluded, both clients SHALL NOT use mixed one bytes and two bytes extensions in the same RTP stream but MAY use one-byte or two-bytes form exclusively (see section 4.1.2).

7. Offer/Answer

The simple signaling described above for the extmap attribute MAY be enhanced in an offer/answer context, to permit:

- o asymmetric behavior (extensions sent in only one direction),
- o the offer of mutually exclusive alternatives, or
- o the offer of more extensions than can be sent in a single session.

A direction attribute MAY be included in an extmap; without it, the direction implicitly inherits, of course, from the stream direction, or is "sendrecv" for session-level attributes or extensions of "inactive" streams. The direction MUST be one of "sendonly", "recvonly", "sendrecv", or "inactive" as specified in [RFC3264]

Extensions, with their directions, MAY be signaled for an "inactive" stream. It is an error to use an extension direction incompatible with the stream direction (e.g., a "sendonly" attribute for a "recvonly" stream).

If an offer or answer contains session-level mappings (and hence no media-level mappings), and different behavior is desired for each stream, then the entire set of extension map declarations MAY be moved into the media-level section(s) of the SDP. (Note that this specification does not permit mixing global and local declarations, to make identifier management easier.)

If an extension map is offered as "sendrecv", explicitly or implicitly, and asymmetric behavior is desired, the SDP answer MAY be changed to modify or add direction qualifiers for that extension.

If an extension is marked as "sendonly" and the answerer desires to receive it, the extension MUST be marked as "recvonly" in the SDP answer. An answerer that has no desire to receive the extension or does not understand the extension SHOULD remove it from the SDP answer.

If an extension is marked as "recvonly" and the answerer desires to send it, the extension MUST be marked as "sendonly" in the SDP answer. An answerer that has no desire to, or is unable to, send the extension SHOULD remove it from the SDP answer.

Local identifiers in the valid range inclusive in an offer or answer must not be used more than once per media section (including the session-level section). A session update MAY change the direction qualifiers of extensions under use. A session update MAY add or

remove extension(s). Identifiers values in the valid range MUST NOT be altered (remapped).

Note that, under this rule, the same local identifier cannot be used for two extensions for the same media, even when one is "sendonly" and the other "recvonly", as it would then be impossible to make either of them sendrecv (since re-numbering is not permitted either).

If a party wishes to offer mutually exclusive alternatives, then multiple extensions with the same identifier in the extended range 4096-4351 MAY be offered; the answerer SHOULD select at most one of the offered extensions with the same identifier, and remap it to a free identifier in the valid range, for that extension to be usable.

Similarly, if more extensions are offered than can be fit in the valid range, identifiers in the range 4096-4351 MAY be offered; the answerer SHOULD choose those that are desired, and remap them to a free identifier in the valid range.

An answerer may copy an extmap for an identifier in the extended range into the answer to indicate to the offerer that it supports that extension. Of course, such an extension cannot be used, since there is no way to specify them in an extension header. If needed, the offerer or answerer can update the session to assign a valid identifier to that extension URI.

Rationale: the range 4096-4351 for these negotiation identifiers is deliberately restricted to allow expansion of the range of valid identifiers in future.

Either party MAY include extensions in the stream other than those negotiated, or those negotiated as "inactive", for example, for the benefit of intermediate nodes. Only extensions that appeared with an identifier in the valid range in SDP originated by the sender can be sent.

Example (port numbers, RTP profiles, payload IDs and rtpmaps, etc. all omitted for brevity):

The offer:

```
a=extmap:1 URI-toffset
a=extmap:14 URI-obscore
a=extmap:4096 URI-gps-string
a=extmap:4096 URI-gps-binary
a=extmap:4097 URI-frametype
m=video
a=sendrecv
m=audio
a=sendrecv
```

The answerer is interested in receiving GPS in string format only on video, but cannot send GPS at all. It is not interested in transmission offsets on audio, and does not understand the URI-obscore extension. It therefore moves the extensions from session level to media level, and adjusts the declarations:

```
m=video
a=sendrecv
a=extmap:1 URI-toffset
a=extmap:2/recvonly URI-gps-string
a=extmap:3 URI-frametype
m=audio
a=sendrecv
a=extmap:1/sendonly URI-toffset
```

8. BNF Syntax

The syntax definition below uses ABNF according to [RFC5234]. The syntax element 'URI' is defined in [RFC3986] (only absolute URIs are permitted here). The syntax element 'extmap' is an attribute as defined in [RFC4566], i.e., "a=" precedes the extmap definition. Specific extension attributes are defined by the specification that defines a specific extension name; there can be several.

Name: extmap

Value: extmap-value

Syntax:

extmap-value = mapentry SP extensionname
 [SP extensionattributes]

mapentry = "extmap:" 1*5DIGIT ["/" direction]

extensionname = URI

extensionattributes = byte-string

direction = "sendonly" / "recvonly" / "sendrecv" / "inactive"

URI = <Defined in RFC 3986>

byte-string = <Defined in RFC 4566>

SP = <Defined in RFC 5234>

DIGIT = <Defined in RFC 5234>

9. Security Considerations

This document defines only a place to transmit information; the security implications of each of the extensions must be discussed with those extensions.

Extensions usage is negotiated using [RFC3264] so integrity protection and end-to-end authentication MUST be implemented. The security considerations of [RFC3264] MUST be followed, to prevent, for example, extension usage blocking.

Header extensions have the same security coverage as the RTP header itself. When Secure Real-time Transport Protocol (SRTP) [RFC3711] is used to protect RTP sessions, the RTP payload can be both encrypted and integrity protected, while the RTP header is either unprotected or integrity protected. In order to prevent DOS attacks, for example, by changing the header extension integrity protection SHOULD be used. Lower layer security protection like DTLS[RFC6347] MAY be used. RTP header extensions can carry sensitive information for which participants in multimedia sessions want confidentiality. RFC6904 [RFC6904] provides a mechanism, extending the mechanisms of SRTP, to selectively encrypt RTP header extensions in SRTP.

Other security options for securing RTP are discussed in [RFC7201].

10. IANA Considerations

This document updates the IANA consideration to reference this document and adds a new SDP attribute in section 10.3

Note to IANA : change RFCxxxx to this RFC number and remove the note.

10.1. Identifier Space for IANA to Manage

The mapping from the naming URI form to a reference to a specification is managed by IANA. Insertion into this registry is under the requirements of "Expert Review" as defined in [RFC5226].

The IANA will also maintain a server that contains all of the registered elements in a publicly accessible space.

Here is the formal declaration to comply with the IETF URN Subnamespace specification [RFC3553].

- o Registry name: RTP Compact Header Extensions
- o Specification: RFC 5285 and RFCs updating RFC 5285.
- o Information required:
 - A. The desired extension naming URI
 - B. A formal reference to the publicly available specification
 - C. A short phrase describing the function of the extension
 - D. Contact information for the organization or person making the registration

For extensions defined in RFCs, the URI SHOULD be of the form urn:ietf:params:rtp-hdext:, and the formal reference is the RFC number of the RFC documenting the extension.

- o Review process: Expert review is REQUIRED. The expert review SHOULD check the following requirements:
 1. that the specification is publicly available;
 2. that the extension complies with the requirements of RTP, and this specification, for header extensions (specifically, that

the header extension can be ignored or discarded without breaking the RTP layer);

3. that the extension specification is technically consistent (in itself and with RTP), complete, and comprehensible;
 4. that the extension does not duplicate functionality in existing IETF specifications (including RTP itself), or other extensions already registered;
 5. that the specification contains a security analysis regarding the content of the header extension;
 6. that the extension is generally applicable, for example point-to-multipoint safe, and the specification correctly describes limitations if they exist; and
 7. that the suggested naming URI form is appropriately chosen and unique.
- o Size and format of entries: a mapping from a naming URI string to a formal reference to a publicly available specification, with a descriptive phrase and contact information.
 - o Initial assignments: none.

10.2. Registration of the SDP extmap Attribute

IANA is requested to register the extmap SDP [RFC4566] attribute.

```
SDP Attribute ("att-field"):  
Attribute name:      extmap  
Long form:          generic header extension map definition  
Type of name:       att-field  
Type of attribute:  Media or session level  
Subject to charset: No  
Purpose:           defines the mapping from the extension  
                  numbers used in packet headers  
                  into extension names.  
Reference:         [RFCXXXX]  
Values:           See [RFCXXXX]
```

10.3. Registration of the SDP extmap-allow-mixed Attribute

The IANA is requested to register one new SDP attribute:

SDP Attribute ("att-field"):
Attribute name: extmap-allow-mixed
Long form: One and Two bytes mixed mode
Type of name: att-field
Type of attribute: Media or session level
Subject to charset: No
Purpose: Negotiate the use of One and Two bytes
in the same RTP stream.
Reference: [RFCXXXX]
Values: None

11. Changes from RFC5285

The major motivation for updating [RFC5285] was to allow having one byte and two bytes RTP header extensions in the same RTP stream (but not in the same RTP packet). The support for this case is negotiated using a new SDP attribute "extmap-allow-mixed" specified in this document.

The other major change is to update the requirement from the RTP specification and [RFC5285] that the header extension "is designed so that the header extension MAY be ignored". This is described in section 4.1.

The transmission consideration section (4.1.1) adds more text to clarify when and how many times to send the RTP header extension to provide higher probability of delivery

>The security section was expanded

The rest of the changes are editorial.

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