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WebRTC Data Channel Protocol
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Abstract

The Web Real-Time Communication (WebRTC) working group is charged to provide protocols to support for direct interactive rich communication using audio, video, and data between two peers' web-browsers. This document specifies an actual (minor) protocol for how the JS-layer dataChannel objects provide the data channels between the peers.

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1. Introduction

The Data Channel Protocol is designed to provide, in the WebRTC context [[I-D.ietf-rtcweb-overview](#)], a generic transport service allowing Web Browser to exchange generic data in a bidirectional peer to peer fashion. As discussed in [[I-D.ietf-rtcweb-data-channel](#)] the protocol uses Stream Control Transmission Protocol (SCTP) [[RFC4960](#)] encapsulated on Datagram Transport Layer Security (DTLS) [[RFC6347](#)] as described in [[I-D.tuexen-tsvwg-sctp-dtls-encaps](#)] to benefit from their already standardized transport and security features.

2. Conventions

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. Terminology

This document uses the following terms:

Association: An SCTP association.

Stream: A unidirectional stream of an SCTP association. It is uniquely identified by a stream identifier.

Channel: A bidirectional channel consisting of two SCTP streams.

4. Opening Handshake

The opening handshake is based on the multimedia session description exchange that happens between the browsers, typically through a Web Server acting as the signaling service.

[[I-D.ietf-mmusic-sctp-sdp](#)] defines the protocol identifier, 'SCTP/DTLS', and defines how to establish an SCTP association over DTLS

using the Session Description Protocol (SDP).

The SCTP association is created with the number of streams specified by the application, and if not specified, then it SHOULD default to 16 streams.

It is recommended that additional streams be available dynamically based on [\[RFC6525\]](#).

[5.](#) Control Messages

Data Channel Control Messages are sent to manage opening bidirectional channels. A `DATA_CHANNEL_OPEN_REQUEST` message is sent on the Stream that is intended to be used to send in that direction, and a response (`DATA_CHANNEL_OPEN_RESPONSE`) is sent back on the Stream to be used for the other direction, with a `reverse_direction_stream` entry holding the Stream number the `DATA_CHANNEL_OPEN_REQUEST` was sent on. This allows association of the Streams that define the bidirectional channel. Finally, a `DATA_CHANNEL_ACK` is sent on the original Stream to complete the 3-way handshake.

Errors are returned by setting the error field of the `DATA_CHANNEL_OPEN_RESPONSE` message to a non-0 value. In this case the original sender of `DATA_CHANNEL_OPEN_REQUEST` shall close the channel.

[5.1.](#) `DATA_CHANNEL_OPEN_REQUEST` Message

This message is sent initially on the stream used for user messages using the channel.

```

      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
+-----+-----+-----+-----+-----+-----+-----+-----+-----+
| Message Type | Channel Type |                   Flags                   |
+-----+-----+-----+-----+-----+-----+-----+-----+-----+
| Reliability Parameter |                   Priority                   |

```

```

+-----+-----+-----+-----+-----+-----+-----+-----+-----+-----+
\                                                                                               /
|                                                                                               |
/                                                                                               \
+-----+-----+-----+-----+-----+-----+-----+-----+-----+-----+

```

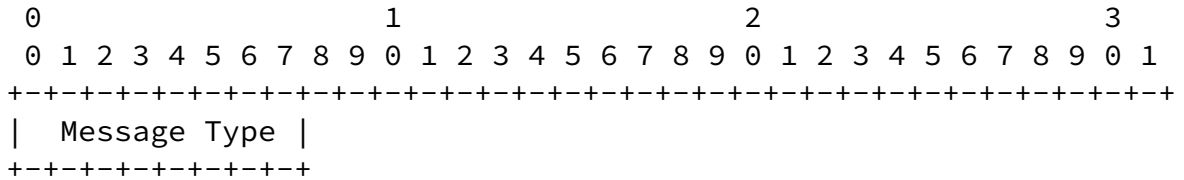
Message Type: 1 byte (unsigned integer)
 This field holds the IANA defined message type for the the DATA_CHANNEL_OPEN_REQUEST message. The suggested value of this field for IANA is 0x00.

Channel Type: 1 byte (unsigned integer)
 This field specifies the type of the channel to be opened:
 DATA_CHANNEL_RELIABLE (0x00): The channel provides a reliable bi-directional communication channel.
 DATA_CHANNEL_RELIABLE_STREAM (0x01): TBD.

DATA_CHANNEL_UNRELIABLE (0x02): TBD.
 DATA_CHANNEL_PARTIAL_RELIABLE_REXMIT (0x03): The channel provides a partial reliable bi-directional Communication channel. User messages will not be retransmitted more times than specified in the Reliability Parameter.
 DATA_CHANNEL_PARTIAL_RELIABLE_TIMED (0x04): The channel provides a partial reliable bi-directional Communication channel. User messages might not be transmitted or retransmitted after a specified life-time given in milli-seconds in the Reliability Parameter. This life-time starts when providing the user message to the Javascript engine.

Flags: 2 bytes (unsigned integer)
 This field contains the bitwise OR of the following flags:
 DATA_CHANNEL_FLAG_OUT_OF_ORDER_ALLOWED (0x0001): If this flag is set, the channel does not need to preserve the message sequencing.

Reliability Parameter: 2 bytes (unsigned integer)
 This field is ignored if a reliable channel is used. If a partial reliable channel with limited number of retransmissions is used, this field specifies the number of retransmissions. If a partial reliable channel with limited lifetime is used, this field specifies the maximum life-time in milli seconds.



Message Type: 1 byte (unsigned integer)
 This field holds the IANA defined message type for the the DATA_CHANNEL_ACK message. The suggested value of this field for IANA is 0x02.

6. Procedures

6.1. Adding a Channel

When one side wants to add a channel, it picks an unused outgoing stream; if no unused streams are available a negotiation to increase the number is done. It then sends a DATA_CHANNEL_OPEN_REQUEST control message on the outgoing stream.

When an DATA_CHANNEL_OPEN_REQUEST is received on an incoming stream, an unused outgoing stream is picked; if no unused streams are available a negotiation to increase the number is done. A DATA_CHANNEL_OPEN_RESPONSE message is sent on the outgoing stream, with the Reverse Stream field set to the incoming stream the DATA_CHANNEL_OPEN_REQUEST message came in on.

When a DATA_CHANNEL_OPEN_RESPONSE is received, the Reverse Stream value is matched against all pending DATA_CHANNEL_OPEN_REQUEST messages. If no match can be found, the DATA_CHANNEL_OPEN_RESPONSE message SHOULD be ignored. If a match is found, then if the error

value is 0 a DATA_CHANNEL_ACK message is sent on the originator's outgoing Stream for the channel. If the error value is non-zero, the open failed, and the originator SHOULD close down the originally-selected outgoing stream and notify the application.

The channel_type and reliability_parameters fields of the DATA_CHANNEL_OPEN_REQUEST message MUST be used to set up the reverse side of the Data Channel so that both directions use the same

options. So both directions are either reliable or use the PR-SCTP extension defined in [RFC3758] using the same policy and parameter.

6.2. Closing a Channel

Data Channels shall be closed by resetting the outgoing stream. If an incoming stream is reset by the peer, a corresponding outgoing stream reset SHOULD be issued. If both streams of a channel are reset, the channel is closed and the streams are available for reuse for new channel opens.

6.3. Sending and Receiving Data

Data shall be sent using PPID's other than the Data Channel Control PPID. These PPID's should be registered with IANA via (TBD). The meaning of these data PPIDs and the format of the data shall be specific to the usage of this protocol, and typically shall be provided to the higher layers to allow proper decoding of the data.

For WebRTC, data PPID's for DOMStrings and binary data blobs shall be created.

All data sent on a Data Channel in both directions MUST be sent over the underlying Stream using the reliability defined when the Data Channel was opened.

Data may be sent before the 3-way handshake is complete; if so it must be sent with in-order delivery set in order to avoid race conditions caused by a handshake message being lost. This is an exception to the requirement to send all data using the channel reliability settings.

It is recommended that message size be kept within certain bounds (TBD).

7. Signaling

This is an application protocol that will normally run on top of SCTP. In the case of rtcweb, it will run over SCTP/DTLS (SCTP on top

the m=application SDP line (Figure 1), details TBD. The <fmt> section should include the virtual port over the rtcweb DataChannel's SCTP association runs on. (This would allow for more than one association to be used on a DTLS connection; it's an open question if this flexibility is needed.)

```
m=application <port> SCTP/DTLS <fmt>
```

Figure 1: Data Channel SDP media line

7.1. Startup considerations

It has been suggested that in order to speed up channel creation that an initial set of channels be allowed to be specified in the SDP, as a very common case for applications will be a fixed set of data channels. This SDP format would support that:

```
m=application <port> SCTP/DTLS <fmt>
a=fmtp:0 channel=1;stream=0;type=reliable;name="foo"
a=fmtp:1 channel=2;stream=1;type=unreliable;inorder;name="bar"
```

Figure 2: Data Channel SDP media plus channels

The answer would echo the channel definitions from the offer, with the associated stream numbers to associated with the channel. This would mean that the data channels would be available to use as soon as the association has been negotiated.

This may need to be specified via MMUSIC.

8. Security Considerations

To be done.

9. IANA Considerations

This document also defines three new SCTP Payload Protocol Identifiers (PPIDs). [RFC 4960](#) [[RFC4960](#)] creates the registry from which these identifiers have been assigned. The following values have been reserved:

WebRTC Control - #To Be Assigned

DOMString - #To Be Assigned
Binary Data - #To Be Assigned

10. Acknowledgments

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