

CLUE Protocol data channel
draft-ietf-clue-datachannel-10

Abstract

This document defines how to use the WebRTC data channel mechanism in order to realize a data channel, referred to as a CLUE data channel, for transporting CLUE protocol messages between two CLUE entities.

The document defines how to describe the SCTPoDTLS association used to realize the CLUE data channel using the Session Description Protocol (SDP), and defines usage of SDP-based "SCTP over DTLS" data channel negotiation mechanism for establishing a CLUE data channel.

Details and procedures associated with the CLUE protocol, and the SDP Offer/Answer procedures for negotiating usage of a CLUE data channel, are outside the scope of this document.

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[1.](#) Introduction

This document defines how to use the WebRTC data channel mechanism [[I-D.ietf-rtcweb-data-channel](#)] in order to realize a data channel, referred to as a CLUE data channel, for transporting CLUE protocol messages between two CLUE entities.

The document defines how to describe the SCTPoDTLS association [[I-D.ietf-tsvwg-sctp-dtls-encaps](#)] used to realize the CLUE data channel using the Session Description Protocol (SDP) [[RFC4566](#)], and defines usage of the SDP-based "SCTP over DTLS" data channel

negotiation mechanism [[I-D.ietf-mmusic-data-channel-sdpneg](#)]. This includes SCTP considerations specific to a CLUE data channel, the SDP Media Description (m- line) values, and usage of SDP attributes specific to a CLUE data channel.

Details and procedures associated with the CLUE protocol, and the SDP Offer/Answer [[RFC3264](#)] procedures for negotiating usage of a CLUE data channel, are outside the scope of this document.

NOTE: The usage of the data channel Establishment Protocol (DCEP) [[I-D.ietf-rtcweb-data-protocol](#)] for establishing a CLUE data channel is outside the scope of this document.

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 [BCP 14](#), [RFC 2119](#) [[RFC2119](#)].

SCTPoDTLS association refers to an SCTP association carried over an DTLS connection [[I-D.ietf-tsvwg-sctp-dtls-encaps](#)].

WebRTC data channel refers to a pair of SCTP streams over a SCTPoDTLS association that is used to transport non-media data between two entities, as defined in [[I-D.ietf-rtcweb-data-channel](#)].

CLUE data channel refers to a WebRTC data channel [[I-D.ietf-rtcweb-data-channel](#)] realization, with a specific set of SCTP characteristics, with the purpose of transporting CLUE protocol [[I-D.ietf-clue-protocol](#)] messages between two CLUE entities.

CLUE entity refers to a SIP User Agent (UA) [[RFC3261](#)] that supports the CLUE data channel and the CLUE protocol.

CLUE session refers to a SIP session [[RFC3261](#)] between two SIP UAs, where a CLUE data channel, associated with the SIP session, has been established between the SIP UAs.

SCTP stream is defined in [[RFC4960](#)] as a unidirectional logical channel established from one to another associated SCTP endpoint, within which all user messages are delivered in sequence except for those submitted to the unordered delivery service.

SCTP identifier is defined in [[RFC4960](#)] as an unsigned integer, which identifies an SCTP stream.

3. CLUE data channel

3.1. General

This section describes the realization of a CLUE data channel, using the WebRTC data channel mechanism. This includes a set of SCTP characteristics specific to a CLUE data channel, the values of the m-line describing the SCTPoDTLS association associated with the WebRTC data channel, and the usage of the SDP-based "SCTP over DTLS" data channel negotiation mechanism for creating the CLUE data channel.

As described in [[I-D.ietf-rtcweb-data-channel](#)], the SCTP streams realizing a WebRTC data channel must be associated with the same SCTP association. In addition, both SCTP streams realizing the WebRTC data channel must use the same SCTP stream identifier value. These rules also apply to a CLUE data channel.

Within a given CLUE session, a CLUE entity **MUST** use a single CLUE data channel for transport of all CLUE messages towards its peer.

3.2. SCTP Considerations

3.2.1. General

As described in [[I-D.ietf-rtcweb-data-channel](#)], different SCTP options (e.g. regarding ordered delivery), can be used for a data channel. This section describes the SCTP options used for a CLUE data channel. [Section 3.3](#) describes how SCTP options are signalled using SDP.

NOTE: While SCTP allows SCTP options to be applied per SCTP message, [[I-D.ietf-rtcweb-data-channel](#)] mandates that, for a given data channel, the same SCTP options are applied to each SCTP message associated with that data channel.

3.2.2. SCTP Payload Protocol Identifier (PPID)

A CLUE entity **MUST** use the PPID value 51 when sending a CLUE message on a CLUE data channel.

NOTE: As described in [[I-D.ietf-rtcweb-data-channel](#)], the PPID value 51 indicates that the SCTP message contains data encoded in a UTF-8 format. The PPID value 51 does not indicate what application protocol the SCTP message is associated with, only the format in which the data is encoded.

3.2.3. Reliability

The usage of SCTP for the CLUE data channel ensures reliable transport of CLUE protocol [[I-D.ietf-clue-protocol](#)] messages.

A CLUE entity MUST NOT use the partial reliability and limited retransmission SCTP extensions, described in [[RFC3758](#)], for the CLUE data channel.

NOTE: [[I-D.ietf-rtcweb-data-channel](#)] requires the support of the partial reliability extension defined in [[RFC3758](#)]. This is not needed for a CLUE data channel, as messages are required to always be sent reliably. [[I-D.ietf-rtcweb-data-channel](#)] also mandates support of the limited retransmission policy defined in [[RFC7496](#)].

3.2.4. Order

A CLUE entity MUST use the ordered delivery SCTP service, as described in [[RFC4960](#)], for the CLUE data channel.

3.2.5. Stream Reset

A CLUE entity MUST support the stream reset extension defined in [[RFC6525](#)].

The dynamic address reconfiguration extension defined in [[RFC5061](#)] MUST be used to signal the support of the stream reset extension defined in [[RFC6525](#)]. Other features of [[RFC5061](#)] MUST NOT be used.

3.2.6. SCTP Multihoming

SCTP multi-homing is not supported for SCTPoDTLS associations, and can therefore not be used for a CLUE data channel.

3.2.7. Close CLUE data channel

As described in [[I-D.ietf-rtcweb-data-protocol](#)], in order to close a data channel, an entity sends an SCTP reset message [[RFC6525](#)] on its outgoing SCTP stream associated with the data channel. When the remote peer receives the reset message, it also sends (unless already sent) a reset message on its outgoing SCTP stream associated with the data channel. The SCTPoDTLS association, and other data channels established on the same association, are not affected by the SCTP reset messages.

3.3. SDP Considerations

3.3.1. General

This section defines how to construct the SDP Media Description (m-line) for describing the SCTPoDTLS association used to realize a WebRTC data channel. The section also defines how to use the SDP-based "SCTP over DTLS" data channel negotiation mechanism [I-D.ietf-mmusic-data-channel-sdpneg] for establishing a CLUE data channel on the SCTPoDTLS association.

NOTE: Other protocols than SDP for negotiating usage of an SCTPoDTLS association for realizing a WebRTC data channel are outside the scope of this specification.

[I-D.ietf-clue-signaling] describes the SDP Offer/Answer procedures for negotiating a CLUE session, including the CLUE controlled media streams and the CLUE data channel.

3.3.1.1. SDP Media Description Fields

As defined in [I-D.ietf-mmusic-sctp-sdp], the field values of an m-line describing an SCTPoDTLS association are set as following:

| media | port | proto | fmt |
|---------------|----------------|-----------------|-------------------|
| "application" | UDP port value | "UDP/DTLS/SCTP" | application usage |
| "application" | TCP port value | "TCP/DTLS/SCTP" | application usage |

Table 1: SDP "proto" field values

CLUE entities SHOULD NOT transport the SCTPoDTLS association used to realize the CLUE data channel over TCP (using the "TCP/DTLS/SCTP" proto value), unless it is known that UDP/DTLS/SCTP will not work (for instance, when the the Interactive Connectivity Establishment (ICE) mechanism [RFC5245] is used and the ICE procedures determine that TCP transport is required).

As defined in [I-D.ietf-mmusic-sctp-sdp], when the SCTPoDTLS association is used to realize a WebRTC data channel, the value of the application usage part is 'webrtc-datachannel'.

3.3.1.2. SDP sctp-port Attribute

As defined in [[I-D.ietf-mmusic-sctp-sdp](#)], the SDP sctp-port attribute value is set to the SCTP port of the SCTPoDTLS association. A CLUE entity can choose any valid SCTP port value.

3.3.2. SDP dcmap Attribute

The values of the SDP dcmap attribute [[I-D.ietf-mmusic-data-channel-sdpneg](#)], associated with the m- line describing the SCTPoDTLS association used to realize the WebRTC data channel, are set as following:

| stream-id | subprotocol | label | ordered | max-retr | max-time |
|--|-------------|----------------------|---------|----------|----------|
| Value of the SCTP stream used to realize the CLUE data channel | "CLUE" | Application specific | "true" | N/A | N/A |

Table 2: SDP dcmap attribute values

NOTE: As CLUE entities are required to use ordered SCTP message delivery, with full reliability, according to the procedures in [[I-D.ietf-mmusic-data-channel-sdpneg](#)] the max-retr and max-time attribute parameters are not used when negotiating CLUE data channels.

3.3.3. SDP dcsa Attribute

The SDP dcsa attribute [[I-D.ietf-mmusic-data-channel-sdpneg](#)] is not used when establishing a CLUE data channel.

3.3.4. Example

```
m=application 54111 UDP/DTLS/SCTP webrtc-datachannel
a=sctp-port: 5000
a=dcmap:2 subprotocol="CLUE";ordered=true
```

Figure 1: SDP Media Description for a CLUE data channel

4. Security Considerations

This specification does not introduce new security considerations, in addition to those defined in [[I-D.ietf-rtcweb-data-channel](#)] and [[I-D.ietf-rtcweb-data-protocol](#)]. Security considerations associated with the CLUE protocol are defined in [[I-D.ietf-clue-protocol](#)].

5. IANA Considerations

5.1. New WebRTC data channel Protocol Value

[RFC EDITOR NOTE: Please replace RFC-XXXX with the RFC number of this document.]

This document adds the 'CLUE' value to the "WebSocket Subprotocol Name Registry" as follows:

| | |
|--------------------------|----------|
| Subprotocol Identifier: | CLUE |
| Subprotocol Common Name: | CLUE |
| Subprotocol Definition: | RFC-XXXX |

6. Acknowledgments

Thanks to Paul Kyzivat, Christian Groves and Mark Duckworth for comments on the document.

7. Change Log

[RFC EDITOR NOTE: Please remove this section when publishing]

Changes from [draft-ietf-clue-datachannel-09](#)

- o Reference updates:
- o - [draft-ietf-tsvwg-sctp-prpolicies](#) published as [RFC 7496](#)
- o - Reference update of draft versions

Changes from [draft-ietf-clue-datachannel-08](#)

- o Changes based on WGLC comments from Daniel Burnett:

- o - Editorial corrections.
- o Changes based on WGLC comments from Paul Kyzivat:
- o - Editorial corrections.

Changes from [draft-ietf-clue-datachannel-07](#)

- o Changes based on WGLC comments from Christian Groves:
- o - IANA considerations for dcmmap attribute removed.
- o - Explicit clarification that the dcmmap attribute max-time and max-retr parameters are not used with ordered/reliable transmission of SCTP messages.
- o - Indication that TCP transport should only be used if ICE is used, and if usage of TCP is required by ICE.
- o - Informative reference to ICE added.
- o - Editorial corrections.
- o Changes based on WGLC comments from Mark Duckworth:
- o - Make it more clear that the rules regarding usage of partial reliability and ordered reliability apply to CLUE data channels.
- o Changes based on WGLC comments from Paul Kyzivat:
- o - Clarify that same SCTP options are applied to each SCTP message associated with a given data channel.
- o - Switched location of sections [3.2](#) and [3.3](#).
- o - PPID table removed. Not needed, since only one value is used.
- o - Editorial corrections.

Changes from [draft-ietf-clue-datachannel-06](#)

- o Usage of DCEP removed.

Changes from [draft-ietf-clue-datachannel-05](#)

- o "DTLS/SCTP" split into "UDP/DTLS/SCTP" and "TCP/DTLS/SCTP".
- o Removed note regarding optionality of including the SDP sctp-port attribute.
- o Added definition of 'SCTPoDTLS association' to the Conventions.
- o Reference to [RFC 4566](#) (SDP) added.

Changes from [draft-ietf-clue-datachannel-04](#)

- o Defines DCEP and external SDP negotiation as two separate mechanisms for negotiating a CLUE data channel.
- o Updates based on technical changes in referenced specifications.
- o Reference to [draft-ietf-mmusic-sctp-sdp](#) added.

Changes from [draft-ietf-clue-datachannel-03](#)

- o IANA considerations added.
- o Editorial changes based on comments from Christian Groves.

Changes from [draft-ietf-clue-datachannel-02](#)

- o SDP m- line example fixed.
- o OPEN ISSUE #1 closed.
- o - It was agreed (IETF#91) to use [draft-ejzak-mmusic-data-channel-sdpneg](#), as it was adopted as a WG item in MMUSIC.
- o - Details for [draft-ejzak-mmusic-data-channel-sdpneg](#) usage added.
- o SDP Offer/Answer procedures removed, as they will be defined in the CLUE protocol draft.
- o References updated.

Changes from [draft-ietf-clue-datachannel-01](#)

- o Support of interleaving "MUST"->"SHOULD".
- o Example updated.
- o Reference update.

Changes from [draft-ietf-clue-datachannel-00](#)

- o SDP Offer/Answer procedures structures according to [RFC 3264](#).
- o Reference update.

Changes from [draft-holmberg-clue-datachannel-04](#)

- o Draft submitted as [draft-ietf-clue-data-channel-00](#).
- o Editorial nits fixed.
- o Changes based on comments from Paul Kyzivat (<http://www.ietf.org/mail-archive/web/clue/current/msg03559.html>).
- o - Proto value fixed.
- o - Explicit text that the partial reliability and limited retransmission policies MUST NOT be used.
- o - Added open issue on whether the DCEP 'protocol' field value for CLUE should contain a version number.
- o - Removed paragraph saying that an offerer must not insert more than one m- line describing an SCTPoDTLS association to be used to realize a CLUE data channel, as the draft already states that only one CLUE data channel per CLUE session shall be opened.
- o - Added reference to [draft-ietf-rtcweb-data-protocol](#) regarding details on resetting SCTP streams.
- o - Added text saying that the value of the DCEP 'channel type' MUST be DATA_CHANNEL_RELIABLE.
- o - Clarified that DCEP must be supported, and used in the absence of another mechanism for opening a CLUE data channel.

Changes from [draft-holmberg-clue-datachannel-03](#)

- o Procedures updated, based on WG agreement (IETF#89) to use DCEP for the CLUE data channel.

- o Procedures updated, based on WG agreement (IETF#89) that offerer is responsible for sending DCEP DATA_CHANNEL_OPEN.
- o Editorial changes, and alignments caused by changes in referenced specifications.

Changes from [draft-holmberg-clue-datachannel-02](#)

- o PPID value for CLUE messages added
- o References updated

Changes from [draft-holmberg-clue-datachannel-01](#)

- o More text added

Changes from [draft-holmberg-clue-datachannel-00](#)

- o Editorial corrections based on comments from Paul K

[8. References](#)

[8.1. Normative References](#)

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