

**CLUE Protocol Data Channel**  
**draft-ietf-clue-datachannel-05**

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 SDP, and defines usage of two mechanisms for establishing a CLUE data channel: the Data Channel Establishment Protocol (DCEP) and the SDP-based "SCTP over DTLS" data channel negotiation mechanism.

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 SDP, and defines usage of two mechanisms for establishing a CLUE data channel: the Data Channel Establishment Protocol (DCEP) [[I-D.ietf-rtcweb-data-protocol](#)] and the SDP-based "SCTP over DTLS" data channel negotiation mechanism [[I-D.ejzak-mmusic-data-channel-sdpneg](#)]. This includes SCTP considerations specific to a CLUE data channel, the SDP Media Description (m-line) values, usage of SDP attributes and DCEP considerations (when DCEP is used) 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.

Simultaneous usage of DCEP and the SDP-based "SCTP over DTLS" data channel negotiation mechanism for opening a CLUE data channel is outside the scope of this specification.

## 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](#)].

WebRTC Data Channel refers to a SCTPoDTLS association [[I-D.ietf-tsvwg-sctp-dtls-encaps](#)] that is used to transport non-media data between two entities, according to the procedures 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 to SIP UAs, where a CLUE data channel, associated with the SIP session, has been established between the SIP UAs.

[RFC4960] defines an SCTP stream 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.



[RFC4960] defines an SCTP identifier as a unsigned integer, which identifies a SCTP stream.

### **3. CLUE Data Channel**

#### **3.1. General**

This section describes the realization of a CLUE Data Channel. 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 either DCEP or 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. SDP Considerations**

##### **3.2.1. 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 construct the SDP dcmap attribute, when the SDP-based "SCTP over DTLS" data channel negotiation mechanism is used for establishing a CLUE Data Channel on the SCTPoDTLS association.

NOTE: Other protocols than SDP for negotiating usage of a 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.

##### **3.2.1.1. 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	"DTLS/SCTP"	application usage

Table 1: SDP "proto" field values

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'.

### 3.2.1.2. 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.

NOTE: Usage of the SDP sctp-port attribute is optional. As defined in [[I-D.ietf-mmusic-sctp-sdp](#)], if the attribute is not present then the default value is 5000.

### 3.2.2. SDP dcpmap Attribute

If the the SDP-based "SCTP over DTLS" data channel negotiation mechanism is used to establish a CLUE data channel, the values of the SDP dcpmap attribute [[I-D.ejzak-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	N/A	N/A	N/A

Table 2: SDP dcpmap attribute values





### **3.2.3. SDP dcsa Attribute**

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

### **3.2.4. Example**

```
m=application 54111 DTLS/SCTP webrtc-datachannel
a=dcmap:2 subprotocol="CLUE"
```

Figure 1: SDP Media Description for a CLUE Data Channel

## **3.3. DCEP Considerations**

### **3.3.1. General**

This section describes how to realize a CLUE data channel using DCEP.

### **3.3.2. Open CLUE Data Channel**

Once the SCTPoDTLS association, used to realize a WebRTC Data Channel has been established, the offerer [[RFC3264](#)] is responsible for establishing the CLUE data channel. The offerer MUST send a DCEP DATA\_CHANNEL\_OPEN message [[I-D.ietf-rtcweb-data-protocol](#)]. The value of the 'protocol' field MUST be "CLUE". The value of the 'channel type' MUST be 'DATA\_CHANNEL\_RELIABLE'.

Once the offerer has received the associated DCEP DATA\_CHANNEL\_ACK message [[I-D.ietf-rtcweb-data-protocol](#)], the CLUE data channel has been established.

If the offerer receives a DCEP DATA\_CHANNEL\_OPEN message, for the purpose of establishing a CLUE data channel, the offerer MUST reset the SCTP stream, in order to prevent two CLUE data channels from being established within the same CLUE session. The offerer MUST NOT send a DCEP DATA\_CHANNEL\_ACK message.

NOTE: If another mechanism than SDP Offer/Answer is used to negotiate the SCTPoDTLS association used to realize the WebRTC Data Channel, that mechanism needs to describe which endpoint is responsible for sending the DCEP\_CHANNEL\_OPEN message, etc.



### 3.3.3. Close CLUE Data Channel

DCEP [[I-D.ietf-rtcweb-data-protocol](#)] does not define a message for closing individual data channels. As described in [[I-D.ietf-rtcweb-data-protocol](#)], in order to close a data channel, a SCTP reset message is sent, in order to close the 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 message.

## 3.4. Sctp Considerations

### 3.4.1. Sctp Payload Protocol Identifier (PPID)

As described in [[I-D.ietf-rtcweb-data-protocol](#)], the PPID value 50 is used when sending a DCEP message on a SCTPoDTLS association used to realize a WebRTC Data Channel.

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.

+-----+-----+		
Protocol	PPID Value	
+-----+-----+		
DCEP	50	
CLUE	51	
+-----+-----+		

Table 3: CLUE Data Channel PPID Values

### 3.4.2. 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 extensions defined in [[RFC3758](#)].

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 [\[I-D.ietf-tsvwg-sctp-prpolicies\]](#).

#### **[3.4.3.](#) Order**

A CLUE entity MUST use the ordered delivery SCTP service, as described in [section 6.6 of \[RFC4960\]](#).

#### **[3.4.4.](#) 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.

#### **[3.4.5.](#) SCTP Multihoming**

SCTP multi-homing is not supported for SCTPoDTLS associations, and can therefor not be used 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



## 5.2. New SDP dcmap attribute subprotocol value

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

OPEN ISSUE: [[I-D.ejzak-mmusic-data-channel-sdpneg](#)] has not yet created a registry for new subprotocol values.

## 6. Acknowledgments

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

## 7. Change Log

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

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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**[I-D.ejzak-mmusic-data-channel-sdpneg]**

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**8.2. Informative References**

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