

AES standard for network and file transfer of audio - Audio-file transfer and exchange - Part 3: Simple project interchange

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Abstract

This standard provides a convention for expressing edit data in text form in a manner that enables simple and accurate computer parsing while retaining human readability. It also describes a method for expressing time-code information in character notation and simple automation for stereo & surround panning and audio gain.

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Contents

0 Introduction	4
0.1 Rationale for this standard	4
0.2 Conventions used in this standard	5
1 Scope	5
2 Normative references	5
3 Definitions	6
4 Edit decision markup language (EDML)	6
4.1 Description	6
4.2 Text specifications and character set.....	6
4.3 Keywords	7
4.4 EDML data types	8
4.5 EDML usage conventions	10
4.6 Required sections and reserved keywords	11
4.7 Column alignment example	12
5 Time-code character groups	13
5.0 General	13
5.1 Time-code group.....	13
5.2 Sample group.....	16
6 Audio decision list (ADL)	18
6.0 ADL version	18
6.1 List structure	18
6.2 Edit time markers	18
6.3 Sections	18
6.4 ADL event entries	26
7 Edit automation	34
7.0 Introduction	34
7.1 Description	34
7.2 Fader section	34
7.3 Pan section	36
7.4 Mute section.....	39
7.5 Markers section.....	40
8 Reference lists	41
9 Assignment of coding	42
9.1 Resource locators, identifiers, and formats.....	42
9.2 Reserved locations	42
Annex A (Normative) Video frame rate to sampling-frequency ratios	43
Annex B (Informative) Time-code character format	44
Annex C (Informative) Section keyword summary	45
Annex D (Informative) Informative references	46
Annex E (Informative) Zone differences from UTC for some locations	47
Annex F (Normative) Currently approved resource locator, identifier, and format references	48
Annex G (normative) ADL Version	49

Foreword

[This foreword is not a part of *AES standard for network and file transfer of audio – Audio-file transfer and exchange – Part 3: Simple project interchange*, AES31-3-1999.]

In 1997-03, the SC-06-01 Working Group on Audio-File Transfer and Exchange of the SC-06 Subcommittee on Network and File Transfer of Audio replaced SC-02-08 to consider the topic of interchange of audio material in the form of computer file data.

This document is part of a proposed multi-part standard for a simple format for audio project interchange defined in project AES-X67. It describes a generalized means to interchange edit data in the form of human-readable text. It describes a means to communicate timing information to sample accuracy in the form of a text string. It describes a technique for communicating edit information between different types of computer platform using plain text.

The remaining projects at the time of issue of this printing, and their intended parts, are:

Project AES-X69: AES31-1 *Media for Interchange*;
Project AES-X66: AES31-2 *Audio File Format*;
Project AES-X68: AES31-4 *Object Oriented Project Interchange*.

Following the presentation of an earlier version of AES31-3 at the AES 105th Convention in San Francisco, CA, US, the draft was substantially revised by Task Group SC-06-01-C at Oxford, England, UK, in 1999-02. It was completed and issued with a call for comment following the working group meeting in conjunction with the AES 106th Convention in Munich, Germany, 1999-05.

At the time of public approval of the standard, the membership of the task group was A. Bell, D. Brennan, C. Broad, K. Brown, J. Bull, G. Dimino, J. Emmett, B. Fattorini, A. Faust, B. Harris, U. Henry, E. Mcdermid, O. Morgan, J. Nunn, M. Porter, I. Rodrigues, C. Sleight, and M. Yonge.

Mark Yonge, Chair
Brooks Harris, Vice-Chair
SC-06-01, 1999-06

Foreword to second edition

This revision of AES31-3 incorporates previous corrigenda, and includes a number of editorial updates and corrections based on experiences of practical implementation. It has also been extended to include support for higher sampling frequencies and simple gain and surround pan automation.

At the time of revision, the writing group included: J. Bull, R. Caine, A. Faust, C. Garcha, B. Harris, U. Henry, J. Palmer, M. Yonge.

Mark Yonge
Chair, SC-02-08

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0 Introduction

0.1 Rationale for this standard

Time-code character format (TCF) is a modified version of the familiar ASCII time-code display. It supports sample-accurate audio editing while integrating easily with systems and processes requiring conventional time codes. It can carry additional information describing time-base derivation. This generalized specification describes features that may not be applicable in all implementations.

Audio sampling resolution provides sample-accurate positions using a video frame count plus audio sample count scheme. The sample count is appended to the time code together with a sampling frequency indicator. Supported sampling frequencies include the primary audio rates together with the pull-up and pull-down rates required for sampling frequency conversions. Sampling frequency is expressed relative to true seconds.

In addition, TCF provides a means for communicating information about the timing relationships between different source media.

Society of Motion Picture and Television Engineers (SMPTE), European Broadcasting Union (EBU), and film time codes indicate how the HH:MM:SS:FF portion of the time code will be interpreted (30 frames per time base for SMPTE, 25 for EBU, 24 for film).

Time base indicates the actual speed (true hertz) at which the time code and the picture material associated with it is resolved. This has two values, 1 Hz and 1/1,001 Hz.

Video fields and time-code mode indicate a) the video field (field one or two) for National Television Systems Committee (NTSC) or Phase Alternating Line (PAL), and b) the frame-code mode (drop frame, non-drop frame) if the time code is NTSC.

Film framing indicates how film frames are transferred to form the video field sequence.

These enhancements provide an underlying mechanism for a broad range of edit data exchange formats capable of tracking film/video/audio phase relationships and conversions throughout the production process. TCF's similarity to conventional time code means TCF can interchange with conventional time-code formats with little difficulty.

0.2 Conventions used in this standard

0.2.1 Decimal points

According to IEC directives, the comma is used in all text to indicate the decimal point. However, in the specified coding, including the examples shown, the full stop is used as in IEC programming language standards.

0.2.2 Data representation

In this standard, all coding and data representations are printed in an equally spaced font.

0.2.3 Non-printing ASCII characters

Non-printing characters are delimited by angle brackets, as in <CR> for carriage return.

1 Scope

This standard provides a convention for expressing edit data in text form in a manner that enables simple and accurate computer parsing while retaining human readability. It also describes a method for expressing time-code information in character notation. It supports common professional audio sampling frequencies, video frame rates, and film framing. This document addresses the core need of the AES31 series of standards in providing a simple but extensible system for passing audio material between systems.

2 Normative references

The following standards contain provisions which, through reference in this text, constitute provisions of this document. At the time of publication, the editions indicated were valid. All standards are subject to revision, and parties to agreements based on this document are encouraged to investigate the possibility of applying the most recent editions of the indicated standards.

ISO/IEC 646:1991, *Information technology — ISO 7-bit coded character set for information interchange*. International Standards Organisation, Geneva, Switzerland.

ISO/IEC 11578, - *Information technology - Open Systems Interconnection - Remote Procedure Call (RPC) [Annex A: Universal Unique Identifier]*; International Standards Organisation, Geneva, Switzerland.

ISO 8601, *Data elements and interchange formats - Information interchange - Representation of dates and times*; International Standards Organisation, Geneva, Switzerland.

AES31-2, *AES standard on network and file transfer of audio – Audio-file transfer and exchange – File format for transferring digital audio data between systems of different type and manufacture*, Audio Engineering Society, New York, NY., US.