



SLOVENSKI STANDARD SIST EN 62365:2009

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Digital audio - Digital input-output interfacing - Transmission of digital audio over asynchronous transfer mode (ATM) networks (IEC 62365:2009)

Digitalton - Digitale Ein-/Ausgangsschnittstellen - Übertragung von Digitalton über ATM-Netzwerke (IEC 62365:2009)

Audionumérique - Interface numérique d'entrée-sortie - Transmission de l'audionumérique sur les réseaux à mode de transfert asynchrone (ATM) (CEI 62365:2009)

Ta slovenski standard je istoveten z: EN 62365:2009

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33.160.30	Avdio sistemi	Audio systems
35.200	Vmesniška in povezovalna oprema	Interface and interconnection equipment

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EUROPEAN STANDARD
NORME EUROPÉENNE
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EN 62365

July 2009

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English version

**Digital audio -
Digital input-output interfacing -
Transmission of digital audio
over asynchronous transfer mode (ATM) networks
(IEC 62365:2009)**

Audionumérique -
Interface numérique d'entrée-sortie -
Transmission de l'audionumérique
sur les réseaux à mode
de transfert asynchrone (ATM)
(CEI 62365:2009)

Digitalton -
Digitale Ein-/Ausgangsschnittstellen -
Übertragung von Digitalton
über ATM-Netzwerke
(IEC 62365:2009)

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This European Standard was approved by CENELEC on 2009-06-01. CENELEC members are bound to comply with the CEN/CENELEC Internal Regulations which stipulate the conditions for giving this European Standard the status of a national standard without any alteration.

Up-to-date lists and bibliographical references concerning such national standards may be obtained on application to the Central Secretariat or to any CENELEC member.

This European Standard exists in three official versions (English, French, German). A version in any other language made by translation under the responsibility of a CENELEC member into its own language and notified to the Central Secretariat has the same status as the official versions.

CENELEC members are the national electrotechnical committees of Austria, Belgium, Bulgaria, Cyprus, the Czech Republic, Denmark, Estonia, Finland, France, Germany, Greece, Hungary, Iceland, Ireland, Italy, Latvia, Lithuania, Luxembourg, Malta, the Netherlands, Norway, Poland, Portugal, Romania, Slovakia, Slovenia, Spain, Sweden, Switzerland and the United Kingdom.

CENELEC

European Committee for Electrotechnical Standardization
Comité Européen de Normalisation Electrotechnique
Europäisches Komitee für Elektrotechnische Normung

Central Secretariat: Avenue Marnix 17, B - 1000 Brussels

Foreword

The text of document 100/1517/FDIS, future edition 2 of IEC 62365, prepared by technical area 4, Digital systems interfaces and protocols, of IEC TC 100, Audio, video and multimedia systems and equipment, was submitted to the IEC-CENELEC parallel vote and was approved by CENELEC as EN 62365 on 2009-06-01.

This European Standard supersedes EN 62365:2005.

The main changes with respect to EN 62365:2005 are listed below:

- second, third and fourth required formats in 4.3 removed;
- 4.3 reformatted, eliminating Table 2, and subsequent tables renumbered.

The following dates were fixed:

- latest date by which the EN has to be implemented at national level by publication of an identical national standard or by endorsement (dop) 2010-03-01
- latest date by which the national standards conflicting with the EN have to be withdrawn (dow) 2012-06-01

Annex ZA has been added by CENELEC.

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Endorsement notice

The text of the International Standard IEC 62365:2009 was approved by CENELEC as a European Standard without any modification. standards.iteh.ai/catalog/standards/sist/2084ce99-5a39-4bf8-8cea-98346761304d/sist-en-62365-2009

In the official version, for Bibliography, the following note has to be added for the standard indicated:

IEC 60027-2 NOTE Harmonized as EN 60027-2:2007 (not modified).

Annex ZA (normative)

Normative references to international publications with their corresponding European publications

The following referenced documents are indispensable for the application of this document. For dated references, only the edition cited applies. For undated references, the latest edition of the referenced document (including any amendments) applies.

NOTE When an international publication has been modified by common modifications, indicated by (mod), the relevant EN/HD applies.

<u>Publication</u>	<u>Year</u>	<u>Title</u>	<u>EN/HD</u>	<u>Year</u>
IEC 60958-1	- ¹⁾	Digital audio interface - Part 1: General	EN 60958-1	2008 ²⁾
IEC 60958-4	- ¹⁾	Digital audio interface - Part 4: Professional applications (TA4)	EN 60958-4	2003 ²⁾
ITU-T Recommendation I.150	- ¹⁾	B-ISDN asynchronous transfer mode functional characteristics	-	-
ITU-T Recommendation I.363.5	- ¹⁾	B-ISDN ATM Adaptation Layer specification: Type 5 AAL	-	-
ITU-T Recommendation Q.2931	- ¹⁾	Digital Subscriber Signalling System No. 2 - User Network Interface (UNI) layer 3 specification for basic call/connection control	-	-

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¹⁾ Undated reference.

²⁾ Valid edition at date of issue.

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**Digital audio – Digital input-output interfacing – Transmission of digital audio
over asynchronous transfer mode (ATM) networks**

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INTERNATIONAL ELECTROTECHNICAL COMMISSION

**DIGITAL AUDIO –
DIGITAL INPUT-OUTPUT INTERFACING –
TRANSMISSION OF DIGITAL AUDIO OVER
ASYNCHRONOUS TRANSFER MODE (ATM) NETWORKS**

FOREWORD

- 1) The International Electrotechnical Commission (IEC) is a worldwide organization for standardization comprising all national electrotechnical committees (IEC National Committees). The object of IEC is to promote international co-operation on all questions concerning standardization in the electrical and electronic fields. To this end and in addition to other activities, IEC publishes International Standards, Technical Specifications, Technical Reports, Publicly Available Specifications (PAS) and Guides (hereafter referred to as "IEC Publication(s)"). Their preparation is entrusted to technical committees; any IEC National Committee interested in the subject dealt with may participate in this preparatory work. International, governmental and non-governmental organizations liaising with the IEC also participate in this preparation. IEC collaborates closely with the International Organization for Standardization (ISO) in accordance with conditions determined by agreement between the two organizations.
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International Standard IEC 62365 has been prepared by technical area 4: Digital systems interfaces and protocols, of IEC technical committee 100: Audio, video and multimedia systems and equipment.

The text of this standard is based on the following documents:

FDIS	Report on voting
100/1517/FDIS	100/1550/RVD

Full information on the voting for the approval of this standard can be found in the report on voting indicated in the above table.

This publication has been drafted in accordance with the ISO/IEC Directives, Part 2.

The main changes with respect to the previous edition (2004) are listed below.

- Second, third, and fourth required formats in 4.3 removed.
- 4.3 reformatted, eliminating Table 2, and subsequent Tables renumbered.

The committee has decided that the contents of this publication will remain unchanged until the maintenance result date indicated on the IEC web site under "<http://webstore.iec.ch>" in the data related to the specific publication. At this date, the publication will be

- reconfirmed;
- withdrawn;
- replaced by a revised edition, or
- amended.

A bilingual version of this publication may be issued at a later date.

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INTRODUCTION

This International Standard describes means for the transmission of professional audio across digital networks, including metropolitan- and wide-area networks, to provide the best performance with regard to latency, jitter, and other relevant factors.

Current-generation wide-area communications are based on two very similar systems, synchronous optical network (SONET) and synchronous digital hierarchy (SDH), SONET being used in the United States and SDH in Europe. On top of them are run integrated services digital network (ISDN), asynchronous transfer mode (ATM), and Internet protocol (IP).

ISDN provides telephone call connections of a fixed capacity that carry one 8-bit value per 125 μ s; when a call is set up, its route through the system is chosen, and the switches that route the data are configured accordingly. Each link, between switches or between switch and end equipment, is formatted into frames that take 125 μ s to transmit, and each data byte is identified by its position in the frame.

ATM, also called broadband ISDN, provides a service similar to ISDN, but with the capacity of each call being specified by the caller. Links are formatted into cells, which consist of a header and 48 data bytes; the header is typically 5 bytes long, and most of it is taken up with the virtual channel identifier (VCI) that shows to which call the cell belongs. Call set-up, routing, and switching are done in the same way as in ISDN, but with calls not being restricted to 1 byte every 125 μ s.

IP provides a very different service, not designed for continuous media such as audio and video. There is no call set-up, and each packet contains enough information within itself to allow it to be routed to its destination. This means that the header is much larger than in the case of ATM, typically 74 bytes, and packets will also typically be much larger, if only because otherwise the overheads would be excessive. Each packet is liable to be routed separately, so two packets that are part of the same flow may well take different routes. This can mean that the one that was sent first does not arrive first.

For many professional audio applications, a round-trip time from the microphone through the mixing desk and back to the headphones of no more than 3 ms is required. Allowing 0,5 ms each for conversion from analog to digital and back again, it follows that the network connections to and from the mixing desk must have a latency of less than 1 ms each. For distances of more than about 200 km, the transmission delay alone will exceed 1 ms, but within a metropolitan area the transmission delay should be no more than 0,25 ms (equivalent to about 50 km), leaving 0,75 ms for packetization, queuing within switches, and resynchronization within the receiving equipment.

Packetization delays are proportional to the size of the transmission unit (frame, cell, or packet), and resynchronization delays depend on how evenly spaced the transmission units are when they arrive at their destination. Both classes of delay are thus small for ISDN and large for IP. Using the format specified in this standard to carry dual-channel IEC 60958-4 audio with a 48 kHz sampling frequency over ATM results in an inter-cell time of 125 μ s, at which ATM will have similar delays to ISDN. A higher sampling frequency or a larger number of channels would reduce the inter-cell time and hence also the delays.

The queuing time within each ISDN switch is likely to be around one frame time or 125 μ s. The ATM documents limit the queuing time in an ATM switch to approximately the inter-cell time for the call, which, as with the other delays, translates into performance similar to that of ISDN for dual-channel 48 kHz IEC 60958-4 audio and better for higher sampling frequencies or larger numbers of channels.

The queuing time within an IP router for normal, best effort, Internet traffic is unbounded, and if the router is congested, packets may simply be thrown away. Resource reservation protocol (RSVP) (see Annex A) allows capacity to be reserved for a particular traffic flow, but it does