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Radiocommunication Sector of ITU

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(10/2011)

**Multichannel sound technology in home
and broadcasting applications**

BS Series
Broadcasting service (sound)



International
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REPORT ITU-R BS.2159-3

Multichannel sound technology in home and broadcasting applications

(2009-2010-05/2011-10/2011)

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1 Introduction

ITU-R has developed Recommendation ITU-R BS.775 for multichannel stereophonic sound system with and without accompanying picture. Multichannel stereo as well as 2-channel stereo audio services are widely used as part of digital broadcasting services. Recommendation ITU-R BS.775 specifies a hierarchy of compatible multichannel sound systems to enhance the directional stability of the frontal sound image and the sensation of spatial reality (ambience), and each loudspeaker is set at the same height as a listener's ears.

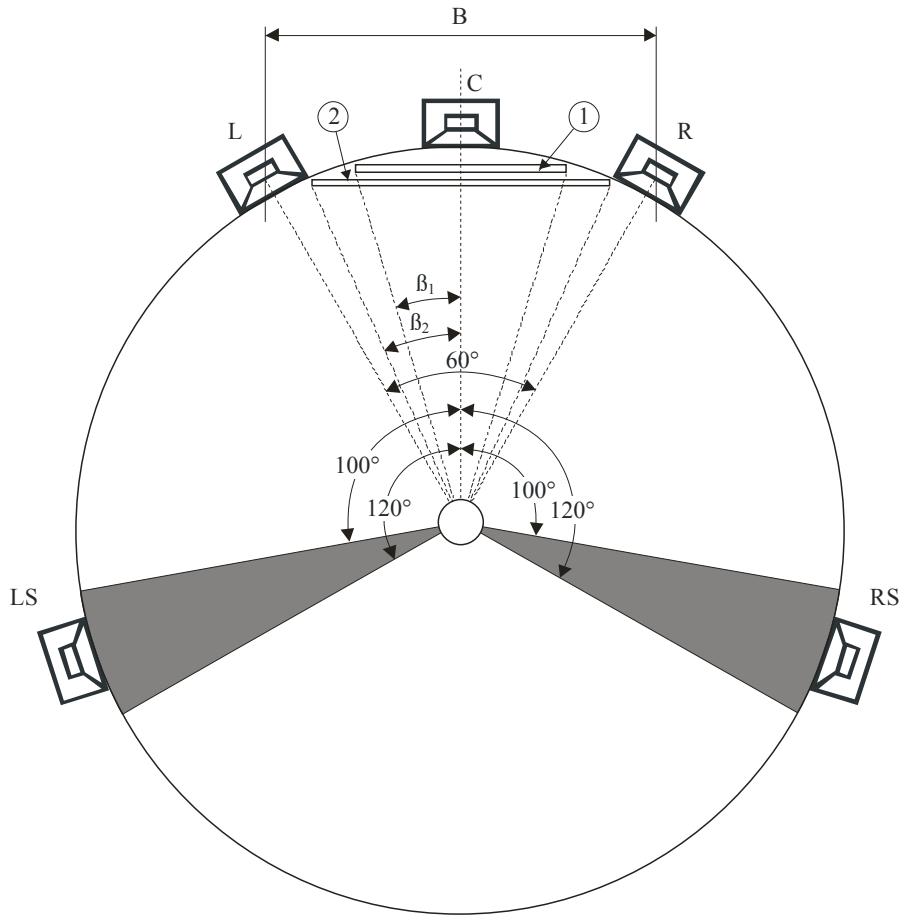
Some television applications with higher resolution imagery and large screen digital imaginary (LSDI) application, both providing wider viewing angle, may need multichannel stereophonic sound systems that can reproduce the sound sources, which are localized at a higher position over the listener and a lower position below the screen, and vertical movements of the sound sources. Several multichannel stereophonic sound systems are currently applied or studied for higher resolution imagery, and some of them have loudspeakers arranged above and below the viewer.

There would be value in continued studies in this area for future broadcasting applications in order to evolve beyond the current 5.1 channel sound system. This Report contains information on the subject of – Multichannel sound technology in home and broadcasting applications, beyond the current 5.1 channel sound system specified in Recommendation ITU-R BS.775.

2 5.1 multichannel sound system

The 5.1 channel sound system has been specified in Recommendation ITU-R BS.775. The system is widely used as a part of digital broadcasting services. It enhances the directional stability of the frontal sound image and the sensation of spatial reality (ambience). The reference loudspeaker arrangement is shown in Fig. 1, in which each loudspeaker is set at the same height as a listener's ears.

FIGURE 1
Reference loudspeaker arrangement with
loudspeakers L/C/R an LS/RS



Screen 1 HDTV - Reference distance = $3 H (2\beta_1 = 33^\circ)$

Screen 2 = $2 H (2\beta_2 = 48^\circ)$

H: height of screen

B: loudspeaker base width

Loudspeaker	Horizontal angle from centre (degrees)	Height (meters)	Inclination (degrees)
C	0	1.2	0
L, R	30	1.2	0
LS, RS	100 ... 120	≥ 1.2	0 ... 15 down

3 Basic requirements of multichannel sound systems beyond the 5.1 sound system

The following requirements are related to the multichannel sound system beyond the current 5.1 channel sound system specified in Recommendation ITU-R BS.775.

1. The directional stability of the frontal sound image should be maintained over the entire higher resolution imagery area. Coincidence of position between sound image and video image also should be maintained over the wide imagery area.
2. The sound image should be reproduced in all directions around the listener, including elevation.
3. The sensation of three-dimensional spatial impression that augments a sense of reality should be significantly enhanced. This may be achieved by the use of side and/or back, top and/or bottom loudspeakers.
4. Exceptional sound quality should be maintained over wider listening area than that provided by current 5.1 channel sound system.
5. Compatibility with the current 5.1 channel sound system specified in Recommendation ITU-R BS.775 should be ensured to an acceptable degree.
6. Live recording, mixing and transmission should be possible.

The reasoning for the basic requirements for advanced multichannel sound systems is provided below:

3.1 Basic requirements of the sound image

- The directional stability of the frontal sound image should be maintained over the entire higher resolution imagery area. Coincidence of position between sound image and video image also should be maintained over the wide imagery area.
- The sound image should be reproduced in all directions around the listener, including in the elevation.

Reason:

The following requirements are defined in Recommendation ITU-R BS.775 for the 5.1 channel sound system:

- *The directional stability of the frontal sound image shall be maintained within reasonable limits over a listening area larger than that provided by a conventional two-channel stereophony.*

The following requirement is also defined:

- *It is not required that the side/rear loudspeakers should be capable of the prescribed image locations outside the range of the front loudspeakers.*

For advanced multichannel sound systems beyond the 5.1 channel sound system, the reproduction of the sound images should be improved from the following two aspects:

- The directional stability of the sound image come from all horizontal directions, i.e. the front/back and left/right directions, should be maintained within reasonable limits over the listening area.
- The sound image included in the elevation directions should also be reproduced.

Therefore, the aforementioned **basic requirement 2)** is defined.

In addition, considering advanced multichannel sound systems applied for television applications, which have high-resolution imagery with a horizontally and vertically wide field of view, the coincidence of the position between sound images and video images is needed over the entire imagery area. Therefore, the aforementioned **basic requirement 1)** is defined.

3.2 Basic requirement of sensation of a spatial impression

- The sensation of three-dimensional spatial impression that augments a sense of reality should be significantly enhanced. This may be achieved by the use of side and/or back, top and/or bottom loudspeakers.

Reason:

The following requirement is defined in Recommendation ITU-R BS.775 for the 5.1 channel sound system:

- *The sensation of spatial reality (ambience) shall be significantly enhanced over that provided by a conventional two-channel stereophony. This shall be achieved by the use of side and/or rear loudspeakers.*

Because each loudspeaker of the 5.1 channel sound system is set at the same height as the listener's ears, the sensation of spatial reality is fundamentally limited to the horizontal plane.

For advanced multichannel sound systems beyond the 5.1 channel sound system, the sensation of spatial impression should be enhanced around the listener, including in the upward/downward elevation sensation, reverberation and ambience. Therefore, the aforementioned **basic requirement 3)** is defined.

3.3 Basic requirement of listening area

- Exceptional sound quality should be maintained over wider listening area than that provided by current 5.1 channel sound system.

Reason:

The following requirement is defined in Recommendation ITU-R BS.775 for the 5.1 channel sound system:

- *The directional stability of the frontal sound image shall be maintained within reasonable limits over a listening area larger than that provided by a conventional two-channel stereophony.*

As frontal two (left and right) loudspeakers are placed for the conventional 2-channel sound system, the listening area of the 5.1 channel sound system should be considered only for the frontal sound image by comparing it to that of a conventional 2-channel stereophony.

To extend the basic requirement of the 5.1 channel sound system, the listening area of advanced multichannel sound systems should be enlarged by comparing them to the 5.1 channel sound system. Therefore, the aforementioned **basic requirement 4)** is defined.

3.4 Basic requirement of compatibility with existing sound systems

- Compatibility with the current 5.1 channel sound system specified in Recommendation ITU-R BS.775 should be ensured to an acceptable degree.

Reason:

The following requirement is defined in Recommendation ITU-R BS.775 for the 5.1 channel sound system:

- *Downward compatibility with sound systems providing lower number of channels (down to stereophonic and monophonic sound systems) shall be maintained.*

The aforementioned compatibility means that, for example, the down-mixed stereophonic or monophonic sound quality from 5.1 channel sound signals should be maintained to an acceptable degree. To extend the basic requirement of the 5.1 channel sound system, the down-mixed 5.1 channel or 2-channel sound quality from advanced multichannel sound signals should be maintained to an acceptable degree for advanced multichannel sound systems.

Additionally, the compatibility should be considered from the view of programme production facilities and exploiting the expertise of sound mixing engineer. Even in future broadcasting services, every programme will not likely be produced by the advanced multichannel sound format. In other words conventional sound formats, such as mono, 2-channel stereo, or 5.1 channel sound format may be operated even in the future broadcasting depending on the programme genre or other service requirements. Thus, broadcasters would prefer to be able to produce various sound programme formats even in a single production studio. As a result, channel compatibility with the 5.1 channel sound system and conventional 2-channel sound system should be considered to an acceptable degree. It also takes advantage of sound mixing engineer's know-how, cultivated by the 5.1 channel sound production. Therefore, the aforementioned **basic requirement 5)** is defined.

3.5 Basic requirement of live broadcasting

- Live recording, mixing and transmission should be possible.

Reason:

The following requirement is defined in Recommendation ITU-R BS.775 for the 5.1 channel sound system:

- *Real-time mixing for live broadcast shall be practicable.*

The live broadcast is the most essential factor for broadcasting services. Therefore, the aforementioned **basic requirement 6)** is defined.

4 Multichannel sound systems beyond the 5.1 sound system under development for broadcasting applications

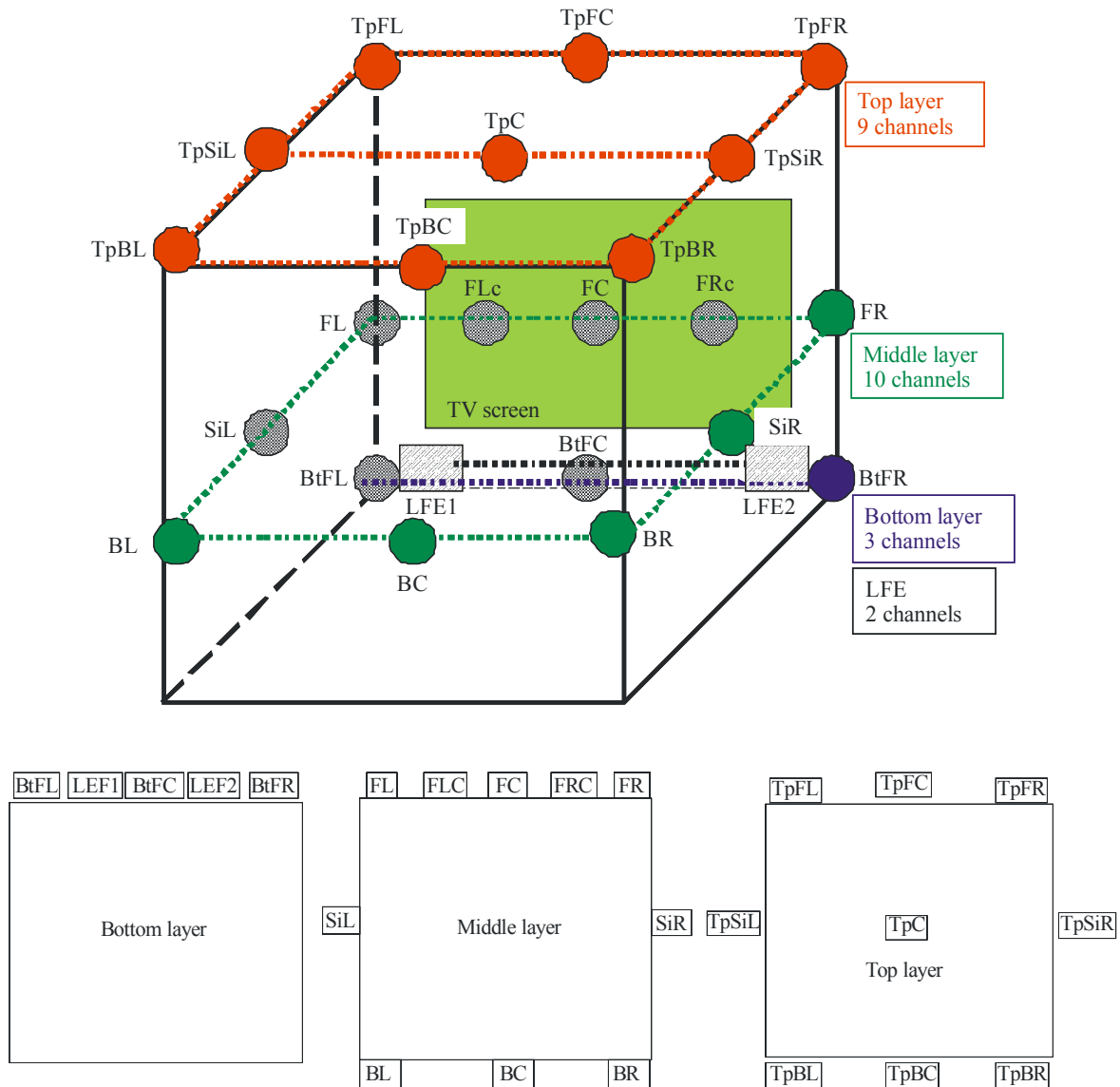
Several multichannel sound systems have been studied to improve the spatial impression of sound. The following systems seem to have the capability for practical use.

4.1 22.2 multichannel sound system

The 22.2 multichannel sound system was developed by NHK (Japan Broadcasting Corporation). It has nine channels at the top layer, ten channels at the middle layer, three channels at the bottom layer and two low frequency effects (LFE) channels. This system is suited to wide screens such as 100 inch FPD display, because it can localize two-dimensionally a sound image over the entire screen by using three bottom channels, five middle channels and three top channels around the screen.

FIGURE 2

The 22.2 multichannel sound system



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This system has common channels at each three layers with other multichannel systems so that its audio can be easily down-mixed to other multichannel sound systems and has compatibility with every multichannel sound system.

Audio characteristics and audio channel mapping for UHDTV programme production for the 22.2 multichannel sound format has been standardized by SMPTE (SMPTE 2036-2), as mentioned in § 8.1.1.

Several sound reproduction systems based on the 22.2 multichannel sound have been developed and exhibited at more than a dozen international expositions and exhibitions, such as World Exposition at Aichi (Japan), NAB Show in Las Vegas (United States of America), IBC in Amsterdam (Netherlands), CEATEC Exhibition in Tokyo (Japan), Broadcast Asia in Singapore, and Grand Exposition for Yokohama's 150th Year.

There are also permanent installations which can reproduce 22.2 multichannel sound. They are:

- One theatrical demonstration room with hundreds of seats in NHK Fureai Hall at Tokyo.
- One theatrical production/demonstration room with hundreds of seats, and one home theatre production/demonstration room in NHK Science & Technology Research Laboratories.
- One laboratory installations in Fraunhofer IIS.
- One laboratory installations in McGill University CIRMMT.

A large theatrical sound system with a 600 inch screen at World Exposition at Aichi, Japan in 2005 is shown in Fig. 3.

FIGURE 3
Theatrical 22.2 multichannel sound system at World Exposition
held at Aichi, Japan in 2005



Home sound systems for 22.2 multichannel sound have also been developed. Figure 4 shows the home 22.2 sound system using multiple compact loudspeakers.

FIGURE 4

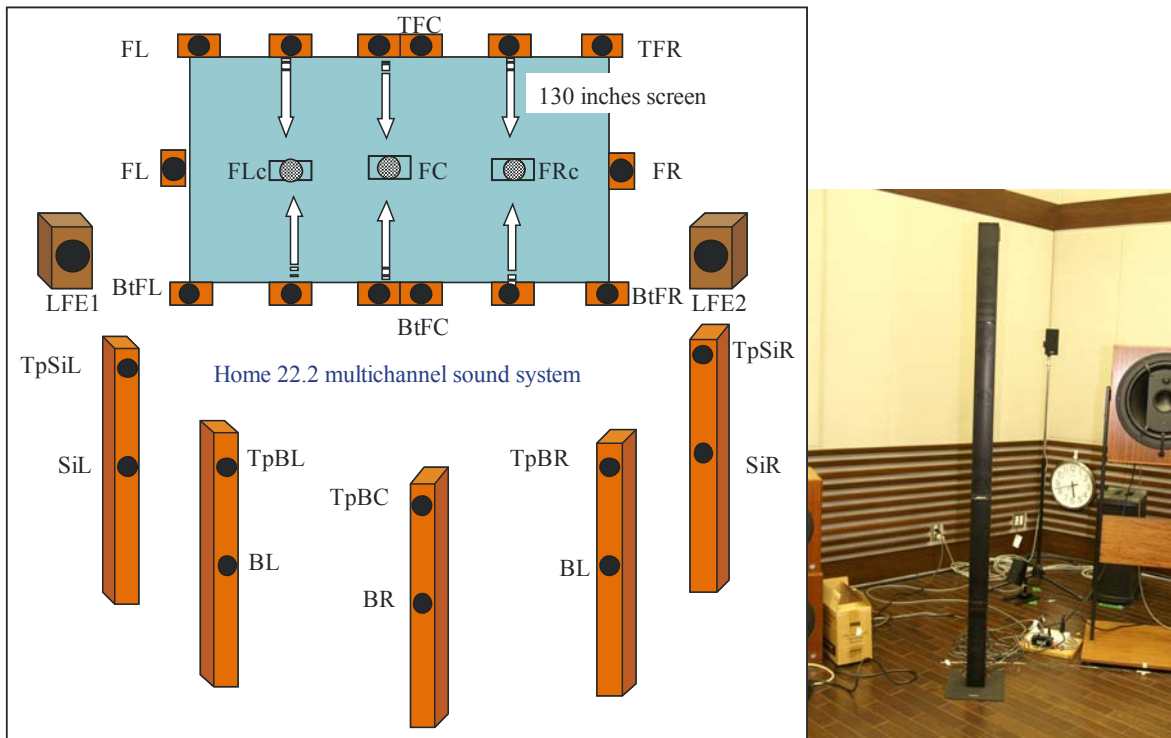
Home 22.2 multichannel sound system using multiple compact loudspeakers

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A tallboy type loudspeaker has been developed to reproduce three vertical channels (i.e. top, middle and bottom channels) by a single loudspeaker. These loudspeakers are used for the home 22.2 multichannel sound system with UHDTV FPD on which compact loudspeaker units are rigged up to reproduce frontal sound channels as Fig. 5.

FIGURE 5

Home 22.2 multichannel sound system using tallboy type loudspeakers



A headphone processor to provide 22.2 multichannel sound has been developed; it is shown in Fig. 6. This processor enables listeners to enjoy an accurate immersive 3D sound with ordinary headphones. Because the headphone processor can reproduce the 22.2 multichannel sound without the use of loudspeakers, TV programmes with 3D sound can be efficiently produced on location in places such as a broadcast OB van.

FIGURE 6
22.2 multichannel sound headphone processor



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4.2 10.2 surround sound system

4.2.1 Background

The Immersive Audio Laboratory is a part of the Integrated Media Systems Center at the University of Southern California and its practitioners have worked since the mid-1990s in the development of multichannel sound, especially 10.2-channel sound. This sound system is a logical extension of Recommendation ITU-R BS.775 and its 5.1-channel layout. Although called 10.2 as shorthand, it actually employs 14 electrical channels, explained below. 10.2 describes the number of loudspeaker locations, since some loudspeaker channels can be combined into one physical location.

4.2.2 Highlights

There are eight permanent installations of 10.2 channel sound as of February 2010. They are:

- Two cinemas with hundreds of seats each. Note that these use a variant on the basic system, designed specifically for cinemas (typ. > 2500 cu. m.), where surround arrays are used for left, right, and rear surround, along with point sources for left and right surround.
- One home theatre demonstration room operating in an audio-video store. In operation for many years and used virtually continuously to demonstrate the advantages of more sound channels to the public.
- One high-power installation at USC's Institute for Creative Technologies, funded by DARPA.
- Two laboratory installations in Ronald Tutor Hall at USC.
- One installation at Inha University, Incheon, Korea.
- One installation in a private home.

In addition, more than a dozen temporary exhibitions have been made on four continents (North America, South America, Europe, and Asia).

There are more than 20 items of produced programme material. Since 10.2-channel sound is a playback platform, not a recording/playback system, a wide variety of methods of recording have

been employed, from classic ones, to completely constructed spaces using advanced digital signal processing algorithms.

The system is scalable from very small listening rooms to cinemas. Changes are made in the physical system to accommodate the range of conditions encountered and its calibration, and the focus of the work has been in deriving the maximum interchangeability among the various size installations. There is no recalibration or mixing necessary to scale from the smallest to the largest space.

The loudspeaker layout was chosen considering physical acoustics of spaces to be reproduced; psychoacoustics of multichannel listening; and the desires of composers, sound designers, and other interested parties. Publications are available detailing these choices.

4.2.3 The loudspeaker channel layout

The loudspeaker channel layout starts with standard 5.1:

- L -30° in plan view, approximately 0° in elevation (raised slightly for line-of-sight in multi-row listening for direct path sound, or the L screen channel in cinemas which are 2/3 of the way up the height of the motion-picture screen to the high-frequency section for instance). Reference point is the centre of the listening area.
- R $+30^\circ$ in plan, same elevation and reference position as L.
- C 0° in plan (straight ahead), same elevation and reference position as L.
- LS direct $-110^\circ \pm 10^\circ$ in plan, same elevation and reference position as L.
- RS direct $+110^\circ \pm 10^\circ$ in plan, same elevation and reference position as L.

To which are added the following:

- Left Wide (LW) -60° in plan, same elevation and reference position as L.
- Right Wide (RW) $+60^\circ$ in plan, same elevation and reference position as L.
- LS diffuse. For “small” rooms of a typical room volume of 85 cu. m: typically a dipole type loudspeaker radiation pattern (low bass excepted) at $-110^\circ \pm 10^\circ$ in plan, elevated above the LS direct loudspeaker. For “large” rooms (cinemas) which are typically $>1\ 000$ cu. m: typically a surround array composed of four to twelve loudspeakers laid out for uniform sound level coverage of the listening area.
- RS diffuse. For “small” rooms as above: typically a dipole-type loudspeaker radiation pattern (low bass excepted) at $+110^\circ \pm 10^\circ$ in plan, elevated above the RS direct loudspeaker. For “large” rooms (cinemas) as above: typically a surround array composed of four to twelve loudspeakers laid out for uniform sound level coverage of the listening area.
- Back Surround (BS): For “small” rooms as above: $+180^\circ$ in plan, same elevation and reference position as L.
- Left Height (LH): -45° in plan and elevated $+45^\circ$ (or whatever is practical) above the listening plane.
- Right Height (RH): $+45^\circ$ in plan and elevated $+45^\circ$ (or whatever is practical) above the listening plane.
- L Sub: Systems employ bass management. Bass below the operating frequency range of all of the left channel loudspeakers (L, LH, LW, LS direct, LS diffuse) and C are added together at equal level and L LFE is added in at $+10$ dB in-band gain. Typical crossover frequency is 25-50 Hz. Typical L LFE low pass filter frequency (brick wall) is 120 Hz. The combined signals are sent to one or more subs located left of the listener. In cinemas they may be in the left front corner. In small rooms they may be on the left side of the room.

- R Sub: Bass below the operating frequency range of all of the right channel loudspeakers (R, RH, RW, RS direct, RS diffuse) and BS are added together at equal level and R LFE is added in at +10 dB in-band gain. Typical crossover frequency is 25-50 Hz. Typical R LFE low pass filter frequency (brick wall) is 120 Hz. The combined signals are sent to one or more subs located right of the listener. In cinemas they may be in the right front corner. In small rooms they may be on the right side of the room.

The consolidated positions of Left Surround direct and diffuse radiators, and Right Surround direct and diffuse radiators (applicable in “small” rooms), result in 10.2 total speaker locations. The system thus requires 14 electrical channels. Additionally, two channels of a sixteen-channel layout are reserved for Hearing Impaired and Visually Impaired descriptive service channels.

4.2.4 Standardization

By following the outlined speaker locations and sound calibration methods, 20 installations have been made to sound as similar as possible. 10.2 is recognized as a format in Apple Quicktime and in SMPTE Digital Cinema standards. It has been implemented by one audio workstation manufacturer, and another is expected to join.

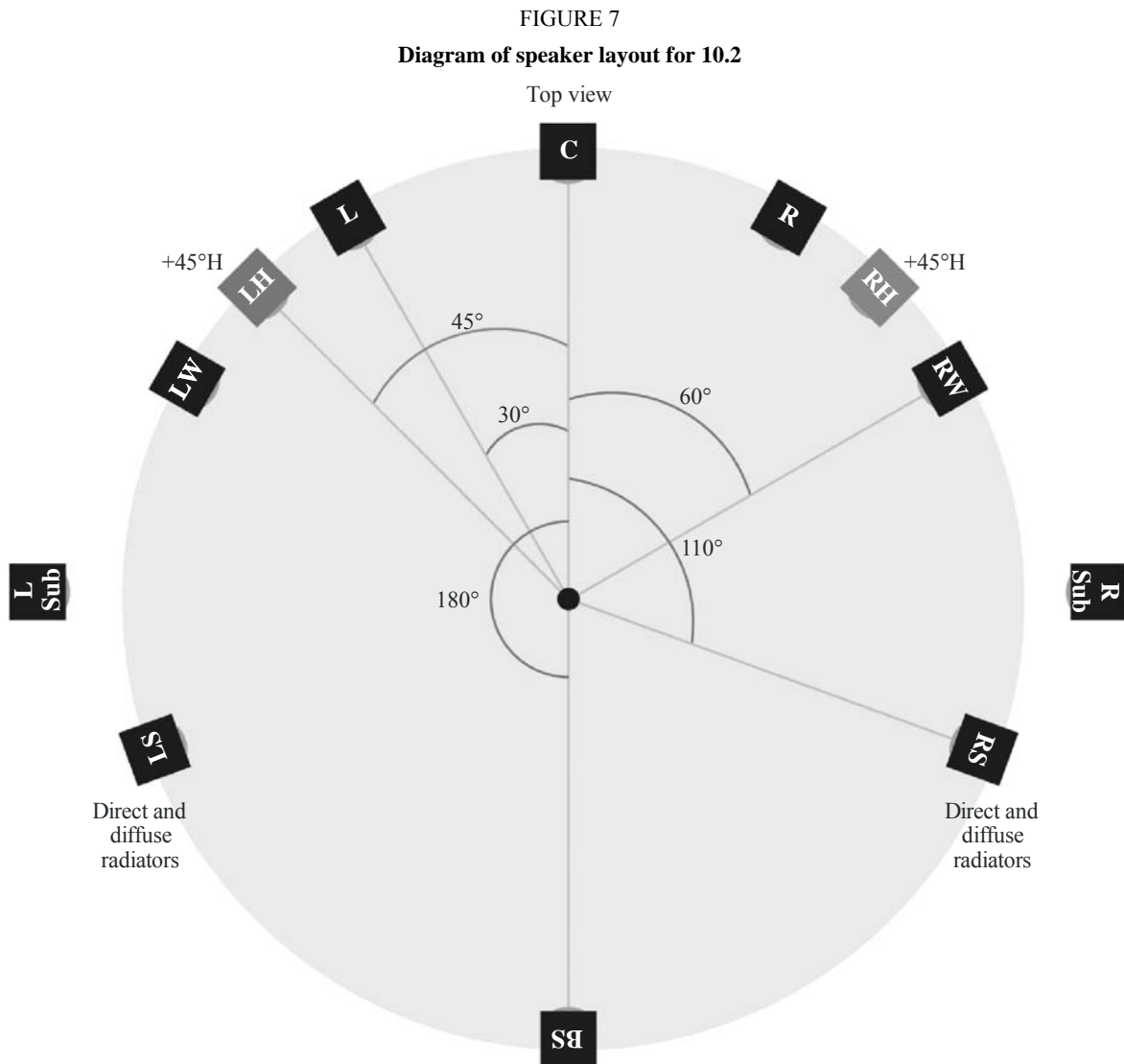
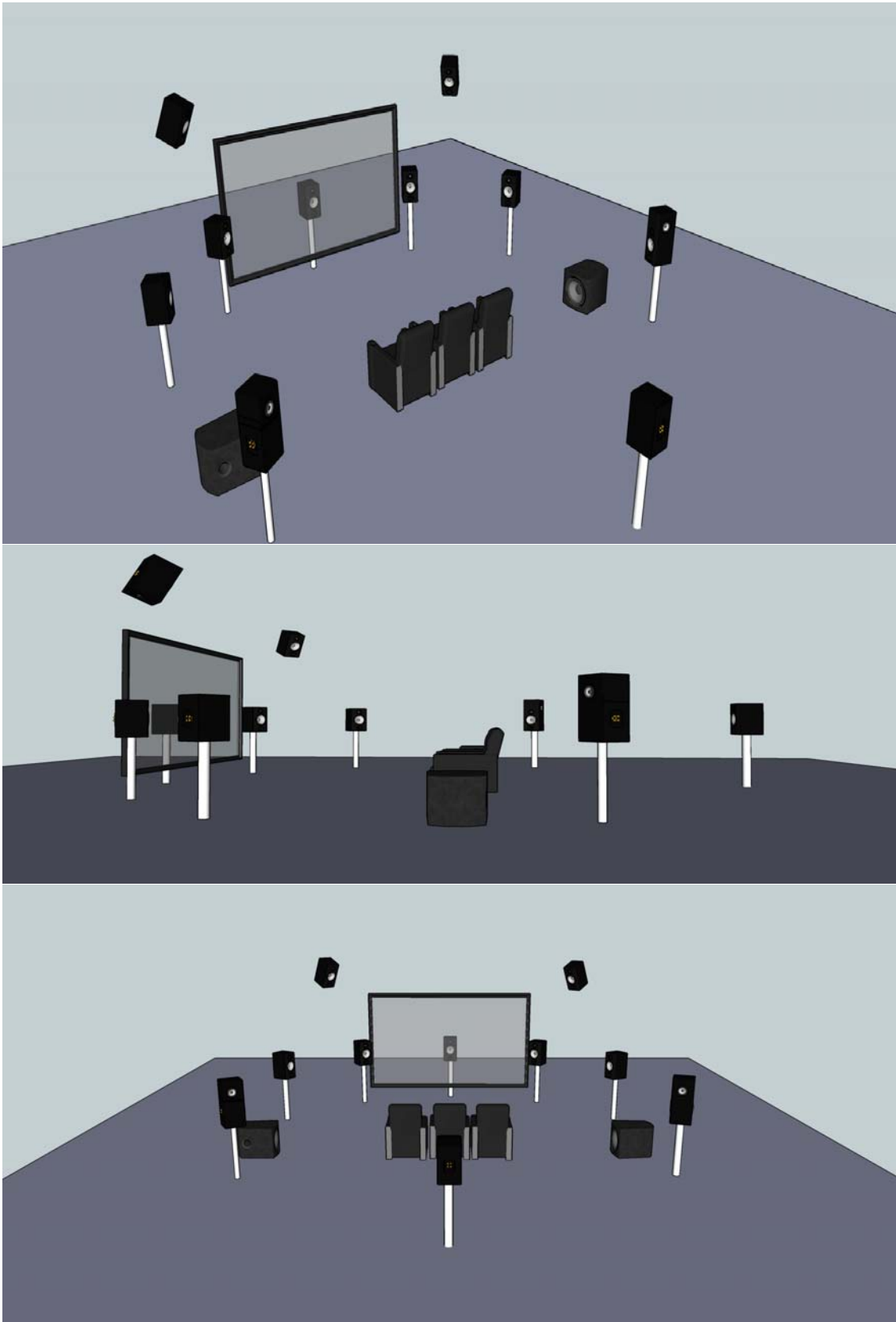


FIGURE 8
Typical “small room” installation schematic

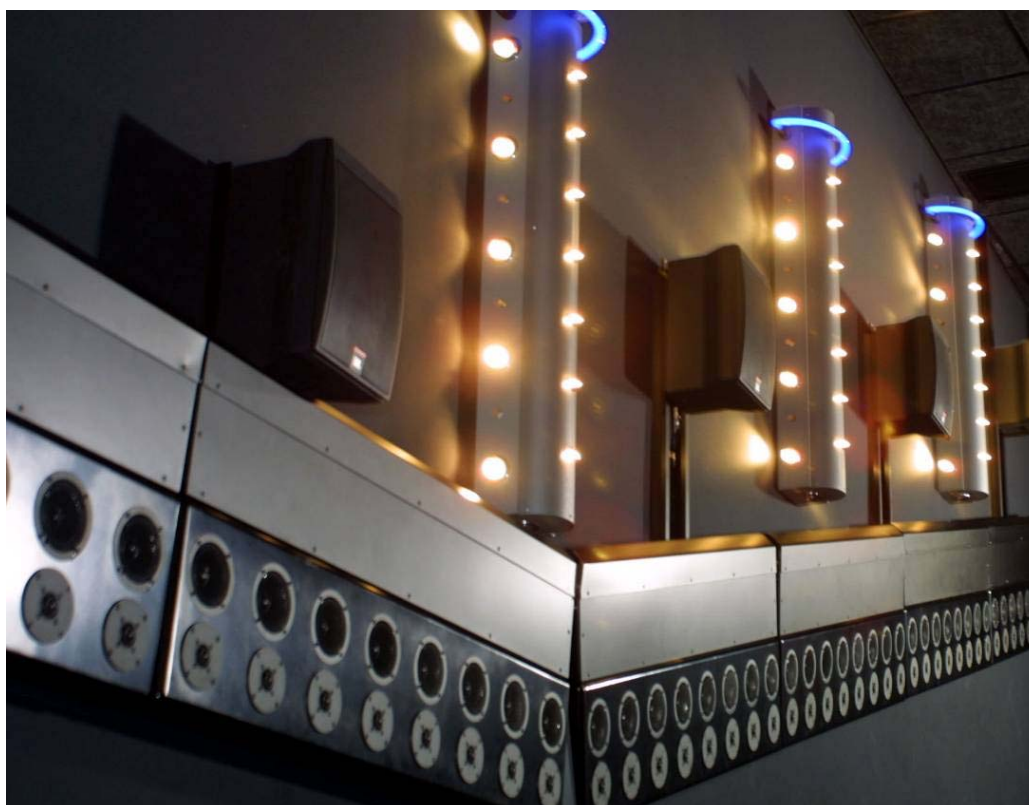


4.3 Wave-field-synthesis

Wave-field-synthesis (WFS) was invented by the Delft University of Technology, Netherlands in 1989. In the European project CARROUSO components for the complete chain including recording, coding, transmission, decoding, and sound reproduction were developed. Since then WFS has been refined to deliver truly immersive sound. Application areas include cinema (with a priority on combination with 3D video), theme parks, virtual reality (VR) installations (in combination with 3D audio) and, in the long run, home theatres. In February 2003 the first cinema using this system started daily operation (Ilmenau, Germany). In 2004 the first WFS system was installed in a sound stage in Studio City, CA. Since 2008, the Chinese 6 Theatre and the Museum of Tolerance in Los Angeles have been equipped with WFS sound systems. These systems are also used in themed environments. Commercial examples of IOSONO GmbH (a spinoff of Fraunhofer IDMT) include the installation in the 4D cinema at the Bavaria Filmstadt (Munich), the Odysseum Science Adventure Park (Cologne), and the “Haunted Mansion” at Disney World (Orlando). Virtual reality installations at the University of Surrey and the Technical University of Ilmenau use WFS with two loudspeaker arrays in the front to enable the proper reproduction of elevation. These two systems also use stereoscopic video projection. An extension of WFS with additional loudspeakers above the listeners was presented at “the 2008 Expo” in Saragossa.

FIGURE 9

WFS sound system in the German cinema “Linden lichtspiele ilmenau”

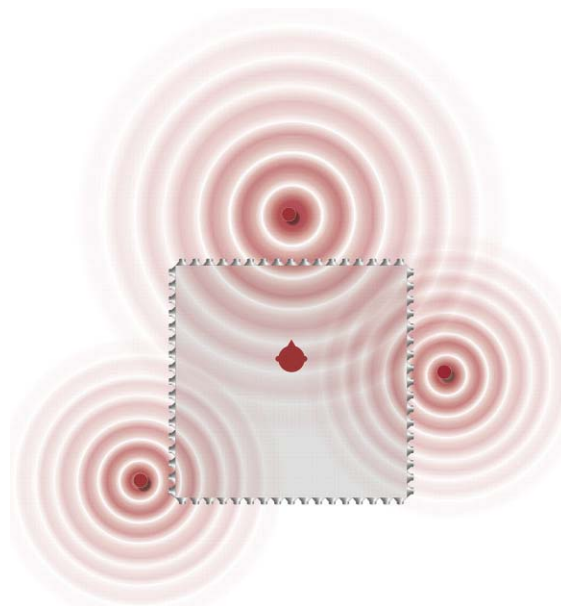


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WFS is an object-oriented approach to accurately recreate a replication of a sound field using the theory of waves and of the generation of wave fronts. This concept is best explained by the well known Huygens principle: points on a wave front serve as individual point sources of spherical secondary waves. This principle is applied in acoustics by using a large number of small and closely

spaced loudspeakers (loudspeaker arrays). The driving signal is calculated for each of the loudspeakers in real time at the reproduction site. The number of loudspeakers is independent from the number of transmission channels and only related to the size of the reproduction room. Loudspeaker arrays controlled by WFS reproduce wave fields that originate from any combination of (virtual) sound sources like an acoustic hologram. When manipulated properly, the system recreates wave fronts approaching perfect temporal, spectral and spatial properties throughout the listening room.

FIGURE 10
Working principle of WFS



Report BS.2159-10

Three representations of sound sources are possible in WFS (see Fig. 11). In the first two, virtual sound sources can be placed behind the loudspeaker arrays (so-called point sources) as well as in front of loudspeaker arrays (so-called focused sources). In the case of sound sources in front of the array, the array radiates convergent waves toward a focus point, from which divergent waves propagate into the listening area. The third type is the so-called plane waves. Plane waves come from the same angular direction for all positions in the listening space.

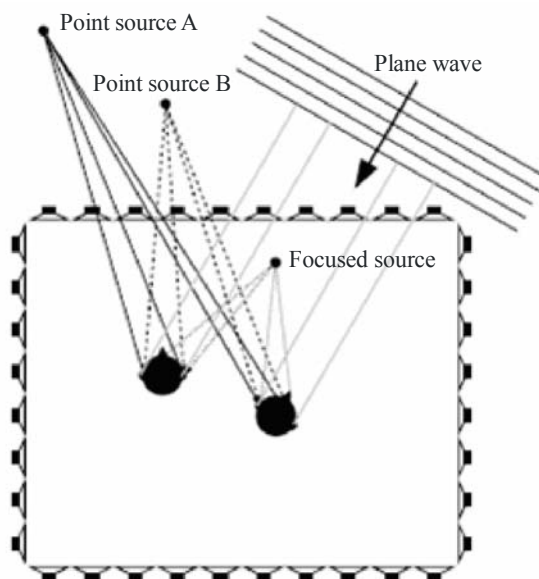
The other commonly used channel-oriented sound reproduction approaches require a well-defined loudspeaker setup, i.e., the number and positions of the loudspeakers are predefined. In the mastering process the target setup must be determined and the loudspeaker signals prepared in a way that allows them to perfectly fit the assumed setup. This implies that it is difficult to feed the generated signals into another sound system.

This problem can be solved by the object-oriented sound reproduction paradigm, which was developed for WFS but which is not restricted to it. In this method, the audio content is represented as audio objects containing the pure audio content together with metadata describing the position of the object in real time along with the properties of the audio object like directivity. On the rendering site the driving signal for each individual loudspeaker is calculated taking into account its exact position in the reproduction room. Besides the positioning of direct sound, a position-dependent calculation of early reflections and diffuse reverberations is possible, which enables the generation of realistic but also artificial spatial environments. Through the availability of the direct sound of

each source and a parametric description of the properties of the room, an optimal reproduction can be adapted to the given spatial environment. This can be a WFS setup of any size (and number of loudspeakers) but also an arbitrary loudspeaker configuration. Increasing the number of loudspeakers increases the size of the sweet spot and makes the sound sources more stable. This results in an increased freedom when deciding which loudspeaker setup to install, because the actual loudspeaker signals are calculated at the reproduction site through a process called rendering.

FIGURE 11

Reproduction of point sources, focused sources and plane waves



Report BS.2159-11

WFS overcomes the restrictions of a sweet-spot and enable the location of sound objects at any position outside and inside the reproduction room without problems of phase or sound coloration. All formats mentioned in §§ 4.1, 4.2 and 5 can be reproduced using WFS by the concept of virtual loudspeakers enabling an enlarged sweet-spot for any content already produced.

4.3.1 Object-based multichannel audio system

This system was developed based on the principles of wave-field synthesis, originally invented by TU Delft and explored in the European project CARROUSO¹. In CARROUSO the MPEG-4 BIFS was used to represent the audio data. For commercial applications this very flexible format proved to be too complex (in terms of storage requirements and computing power) and therefore a less expensive version of the file format had to be developed. With the intention of keeping the perceptual properties of wave-field synthesis, a system for flexible 3D speaker layouts was developed in Germany².

¹ Partners in CARROUSO: Fraunhofer IDMT (Federal Republic of Germany), IRT (Federal Republic of Germany), University of Erlangen (Federal Republic of Germany), France Telecom (France), IRCAM (France), Thales (France), TU Delft (Netherlands), Aristotle University of Thessaloniki (Greece), EPFL (Switzerland), Studer (Switzerland).

² By IOSONO GmbH Erfurt (Federal Republic of Germany) and Fraunhofer IDMT Ilmenau (Federal Republic of Germany).

A complete production and reproduction chain based on the object-based paradigm is available. More than 20 commercial and demonstration installations of systems exist worldwide (e.g. Chinese 1 Multiplex Theater and Chinese 6 Multiplex Theater in Hollywood, Los Angeles). Virtual reality installations at the University of Surrey and the Technical University of Ilmenau use WFS with two loudspeaker arrays in the front to enable the proper reproduction of elevation. These two systems also use stereoscopic video projection. An extension of WFS with additional loudspeakers above the listeners was presented at “the 2008 Expo” in Saragossa. A 3D setup with two layers of loudspeakers was shown at Prolight + Sound 2011 in Frankfurt, Germany. A few installations are shown here to illustrate the diversity that can be realized using the object-based audio system. Content can be exchanged between all these systems.

FIGURE 12

Installation with 64 loudspeakers at the Chinese Multiplex Theater, Hollywood



Report BS.2159-12

FIGURE 13

Setup with 2 flexible layers of 34 loudspeakers presented at a trade show



Report BS.2159-13

FIGURE 14

Wave-field synthesis based setup at Peltz Theatre, Beverly Hills



Report BS.2159-14

A headphone processor to process object-based scene description for dynamic binaural headphone reproduction has been developed. The headphone processor can be used to simulate several loudspeaker layouts to monitor the auditory scene as it would be rendered in a real loudspeaker setup.

4.4 Object-based audio formats

The other commonly used channel-oriented sound reproduction approaches require a well-defined loudspeaker setup, i.e. the number and positions of the loudspeakers are predefined. In the mastering process the target setup must be determined and the loudspeaker signals prepared in a way that allows them to perfectly fit the target setup. This implies that it is difficult to feed the generated signals into another sound system. This problem can be solved by the object-based sound reproduction paradigm, which was developed for WFS but which is not restricted to it.

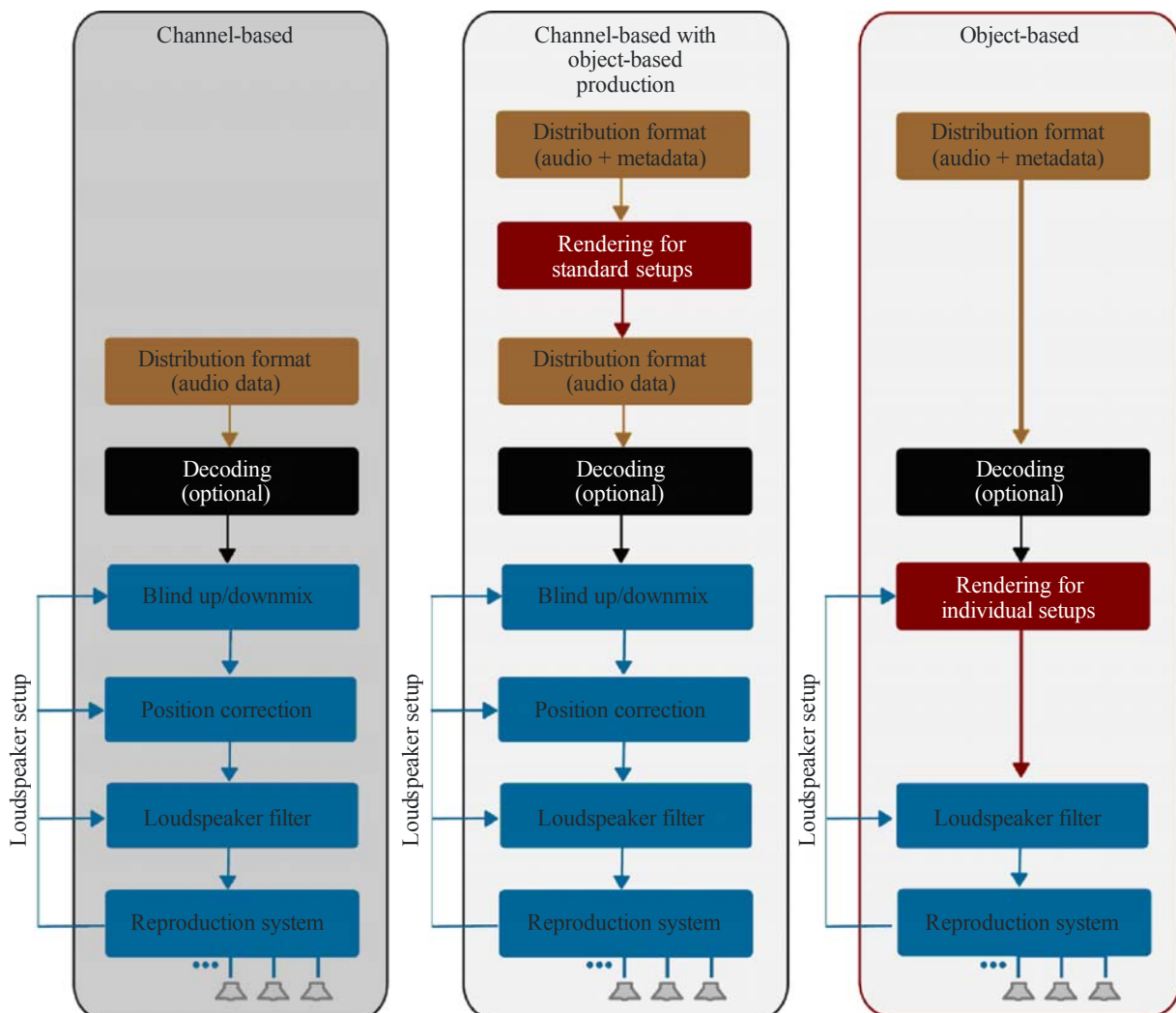
In an object-oriented system the audio content is created independently of any specific loudspeaker layout. The audio content is represented as audio objects containing the pure audio content, together with metadata describing the position of the audio object along with the properties of the audio object such as directivity in real time.

On the rendering site the driving signal for each individual loudspeaker is calculated, taking into account its exact position in the reproduction room. Such representations can be rendered in real time to loudspeaker setups from 5 to more than 500 speakers. The setups do not have to be regular or in a specific layout but standard layouts can easily be supported (as shown in Fig. 15). Furthermore, the auditory scene can be scaled to the current screen size and size of the audience area in a reproduction venue. In that way the content can be transferred between different cinemas as well as to domestic-size screens. Due to the adaptive rendering, loudspeakers do not have to be placed in a specific relationship to the screen. The setup of a system in a home environment becomes flexible and acceptable. This results in an increased freedom when deciding which loudspeaker setup to install, because the actual loudspeaker signals are calculated at the reproduction site through a process called rendering.

WFS overcomes the restrictions of a sweet-spot and enables the location of sound objects at any position outside and inside the reproduction room without problems of phase or sound coloration, if an appropriate number of loudspeakers are installed. All current or future multichannel formats can be reproduced using WFS by the concept of virtual loudspeakers enabling an enlarged sweet-spot for any content already produced.

FIGURE 15

Comparison between channel based and object-based production system



4.4.1 Rendering and reproduction of object-based audio

Depending on the specified speaker setups the algorithm scales the reproduction of an object-based scene. If only a few loudspeakers are available, a rendering with comparable quality to any multichannel format is the result. On the other end, if a wave-field synthesis loudspeaker setup is available, wave-field synthesis is used for the rendering process. Due to its flexibility loudspeaker signals for multichannel layouts like 22.2, 10.2, 9.1 or 5.1 can be rendered in real time directly using the production or reproduction tools. Using the spatial audio processor a rendering with a specific adaptation to a venue is possible and loudspeaker setups from 5 to 500 speakers are possible. Such a system can reproduce different source types which are known from wave-field synthesis. Point sources enable the perception of a fixed source position for the whole audience area. Plane waves enable the perception of a fixed source direction for the whole audience area. Depending on the number of loudspeakers the focusing can be used to create a source position between loudspeaker and listener.

5 Multichannel sound systems in use for home audio release media

The following multichannel sound systems are used in home audio entertainment.

5.1 DVD audio

DVD audio is a digital format for delivering exceptionally high-fidelity audio content on a DVD. It offers many channel configurations of audio channels, ranging from mono to 5.1-channel surround sound, at various sampling frequencies and bit resolution per sample (from compact disc 44.1 kHz/16 bits up to 192 kHz/24 bits). Compared with the CD format, the much higher capacity DVD format enables the inclusion of considerably more music (with respect to total running time and quantity of songs) and/or far higher audio quality (reflected by higher sampling frequencies and greater bit resolution per sample, and/or additional channels for spatial sound reproduction).

Audio is stored on the disc in linear pulse code modulation (PCM) format, which is either uncompressed or losslessly compressed with Meridian Lossless Packing (MLP). The maximum permissible total bit rate is 9.6 Mbit/s. In uncompressed modes, it is possible to get up to 96 kHz/16 bits or 48 kHz/24 bits in 5.1-channel surround sound. To store 5.1-channel surround sound tracks in 88.2 kHz/20 bits, 88.2 kHz/24 bits, 96 kHz/20 bits or 96 kHz/24 bits MLP encoding is mandatory.

5.2 SACD

Super Audio CD (SACD) is a read-only optical audio disc format that provides higher fidelity digital audio reproduction. SACD audio is stored in a format called Direct Stream Digital (DSD), which differs from the conventional PCM used by compact disc or conventional computer audio systems. DSD is 1-bit and has a sampling frequency of 2.8224 MHz. This gives the format a greater dynamic range and wider frequency response than that of the CD. The system is capable of delivering a dynamic range of 120 dB from 20 Hz to 20 kHz and an extended frequency response up to 100 kHz.

SACD supports up to six channels at full bandwidth. In its current form the SACD standard does not precisely specify how the channels shall be used.

222 sound currently uses SACD to provide 2 + 2 + 2 sound contents consist of 6 channels including 4 channels (front left, front right, rear left and rear right) and 2 height channels (top front left and top front right).

5.3 BD

BD is an optical disc format. The format was developed to enable recording, rewriting and playback of high-definition (HD) video, as well as storing large amounts of data. BD pre-recorded application format (BD-ROM) is designed not only for pre-packaged HD movie content but also as a key component of a consumer HD platform. The BD platform is designed to provide access to HD content throughout the home via HD digital broadcast recording and HD playback functions.

One of the key features offered by BD-ROM is

- Industry standard high definition video and surround sound audio:
 - MPEG-2, MPEG-4 AVC, and SMPTE VC-1 video formats;
 - LPCM as well as Dolby Digital, Dolby Digital Plus, Dolby Lossless, DTS digital surround, and DTS-HD audio formats.

BD-ROM supports six types of audio stream formats ranging from a low bit rate to high audio quality, as shown in Table 1.

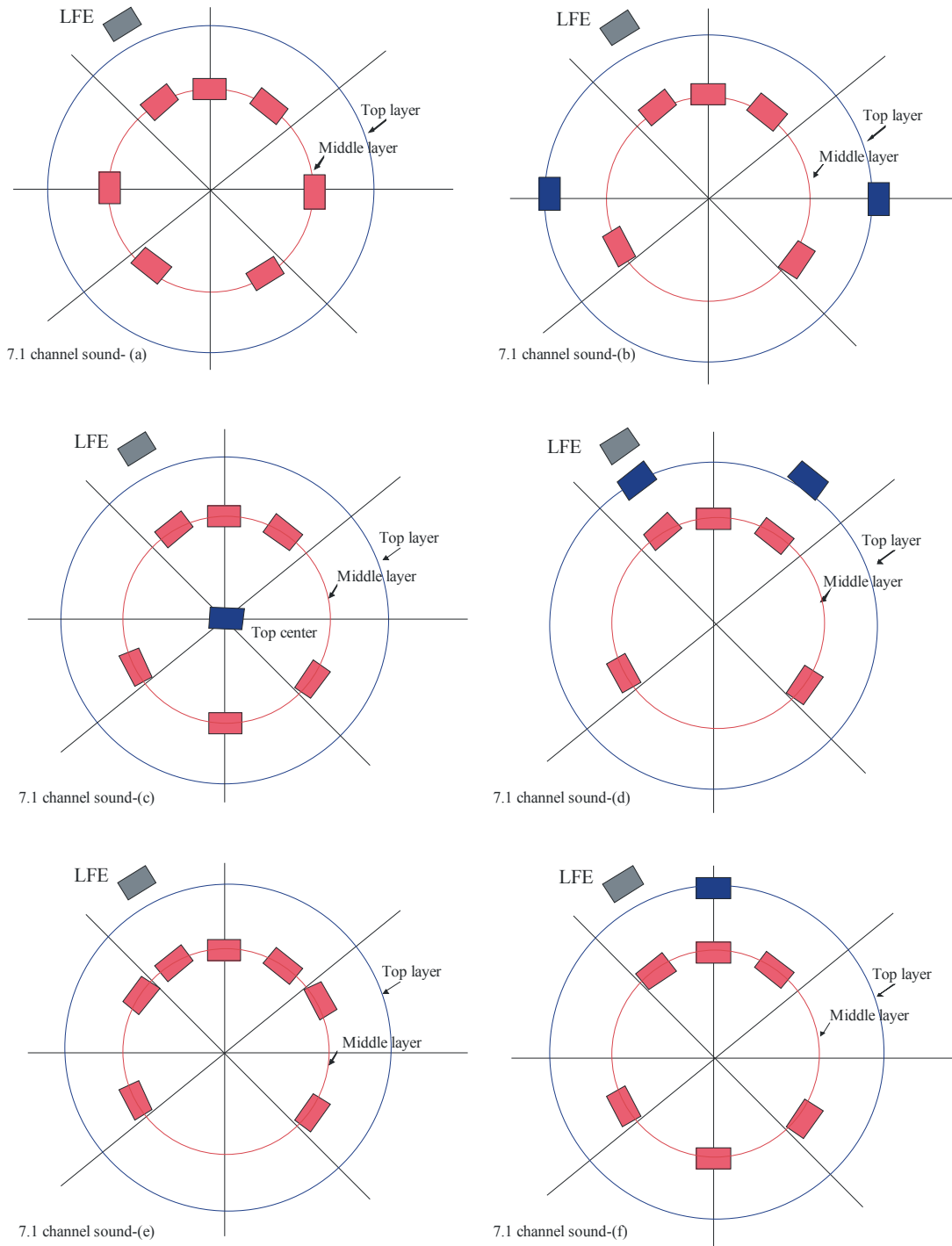
TABLE 1
Specification of BD-ROM audio streams

CODEC	LPCM	Dolby Digital	Dolby Digital Plus	Dolby lossless	DTS digital surround	DTS-HD
Max. bit rate	27.648 Mbit/s	640 kbit/s	4.736 Mbit/s	18.64 Mbit/s	1.524 Mbit/s	24.5 Mbit/s
Max.ch	8(48 kHz, 96 kHz), 6(192 kHz)	5.1	7.1	8(48 kHz, 96 kHz), 6(192 kHz)	5.1	8(48 kHz, 96 kHz), 6(192 kHz)
bits/sample	16, 20, 24	16-24	16-24	16-24	16, 20, 24	16-24
Sampling frequency	48 kHz, 96 kHz, 192 kHz	48 kHz	48 kHz	48 kHz, 96 kHz, 192 kHz	48 kHz	48 kHz, 96 kHz, 192 kHz

Whilst 7.1 channel sound is available in Dolby Digital Plus and DTS-HD, several channel mappings are proposed in terms of 7.1 channel sound as shown in Fig. 16. The proposed mappings consist of two layers of loudspeaker positions, middle and top layer. The middle layer is basically at the same height with the listener's ear and the top layer is at a higher position such as at ceiling level.

FIGURE 16

Examples of loudspeaker mapping of 7.1 channel sound



6 Multichannel sound programme production in studio for home audio

6.1 Production of 5.1, 6.1 and 7.1 channels

Many countries are currently producing 5.1 channel sound programmes for broadcasting and audio and video releases. Production of 6.1 channel and 7.1 channel sound programmes is also increasing for audio and video releases. Several microphone techniques had been already proposed by many sound engineers and audio researchers for 5.1 channel sound recording. As described above, 7.1 channel sound is functional with the loudspeakers at a higher position. Several issues regarding how to use height channel properly or effectively were discussed in various workshops.

6.2 Production of 22.2 multichannel sound

6.2.1 Principles of three-dimensional sound mixing

NHK has already produced several UHDTV programmes with 3D sound using the 22.2 multichannel sound mixing system. Sound engineers and designers have been developing know-how and experience in the 3-D sound field. The current, conventional applications of layers on 22.2 multichannel sound used for mixing are enumerated below.

Top layer

- Reverberation and ambience.
- Sound localized above, such as loudspeakers hung in gymnasiums and airplanes and at fireworks shows.
- Unusual sound, such as meaningless sound.

Middle layer

- Basic sound field formation.
- Envelopment reproduction.

Bottom layer

- Sounds of water such as the sea, rivers, and drops of water.
- Sound on the ground in scenes with bird's-eye views.

Sound engineers have also been discussing several issues in 3D sound mixing. The principal issues are as follows.

- Effective use of the top and bottom layer.
- 3D movement of sound images.
- Creating a sense of elevation.
- Interaction between immersive audio and visual cues.

6.2.2 22.2 multichannel sound post-production system

A 22.2 multichannel sound post-production system has been developing for producing 3D sound. This system currently includes a Digital Audio Workstation and a sound mixing console with 3D pan on each channel strip, 3D audio signal compressor on 24-channel master bus and a down-mixing function. It can mix over 1 000 sound tracks to produce 22.2 multichannel sound.

FIGURE 17

22.2 multichannel sound mixing console

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6.2.3 Examples of three-dimensional sound live mixing**6.2.3.1 A large-scale musical TV programme at NHK Hall**

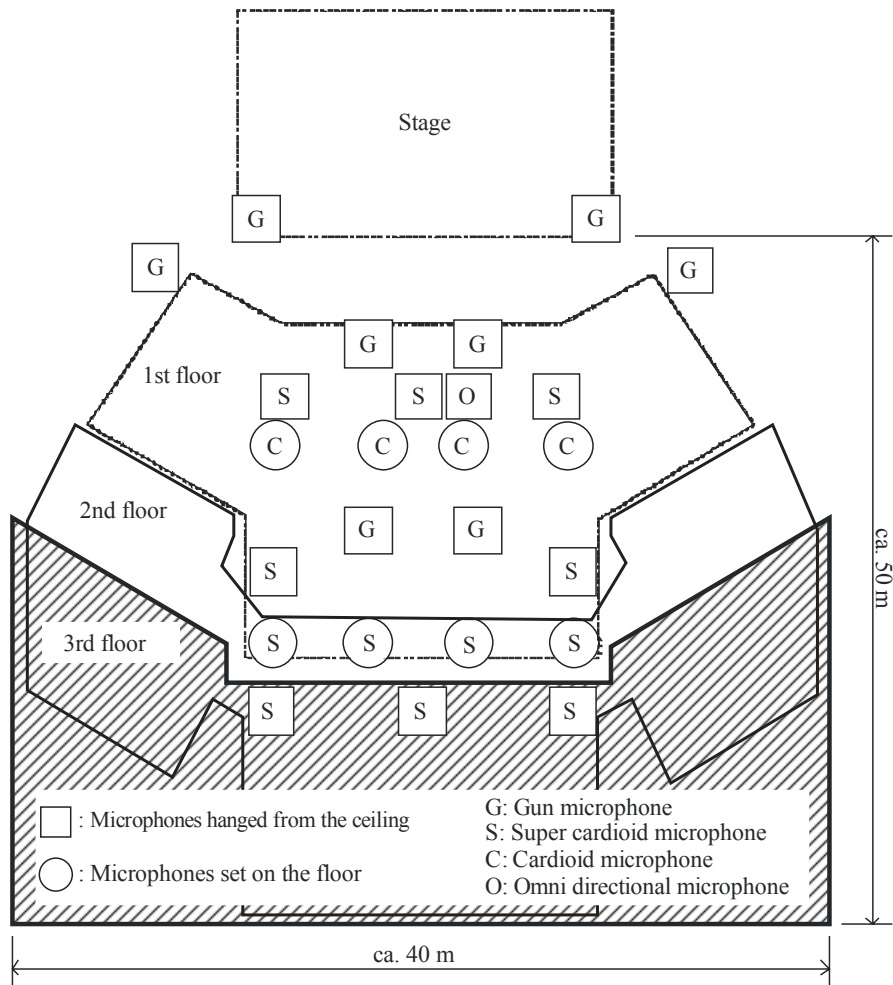
The 22.2 multichannel sound of a large-scale musical TV programme at NHK Hall was live mixed using about 150 sound input feeds. Multiple microphones were arranged in the manner of standard pop recording, basically as a “close setup near the sound sources”, so the 22.2 multichannel sound mixing was also done with the conventional pop music mixing technique, i.e., the multi-microphone recording technique. The major difference in microphone arrangement and mixing between 5.1 channel sound and 22.2 multichannel sound is in how the ambience of a concert hall is recorded and mixed. It is important to reproduce the acoustical impression of the huge dimensions of the NHK Hall, which has a 4 000-seat capacity and the impression of being surrounded by an enormous audience. Spatial sound reproduction advantages of the 22.2 multichannel system include the improvement of the listener’s sense of envelopment and the enlargement of the listening area with exceptional sound quality. For the achievement of these new features of spatial reproduction with a 22.2 multichannel sound system, the following concept was planned as shown in Fig. 18.

- Reflection and reverberation in the auditorium of NHK Hall, which are captured by microphones hung from the ceiling, are reproduced by the top layer loudspeakers of the 22.2 multichannel sound system to widen the listening area and create a sense of the listener being enveloped.
- The sounds of the audience, such as applause and shouts of encouragement, which are captured by several microphones set close to the audience, are reproduced by the middle layer loudspeakers to give the viewers a good sense of presence, as if they were sitting in the audience in NHK Hall.
- As the sound of musical instruments and vocals are reproduced by the sound reinforcement (SR) loudspeaker system in NHK Hall, reproduced sound reflected by the wall, ceiling,

and floor of the hall is captured by the ambience microphones and reproduced by the top and middle layer loudspeakers to give the viewers the same sense of presence.

FIGURE 18

Arrangement of microphone for live mixing of a large-scale musical UHDTV programme by the 22.2 multichannel sound at NHK Hall



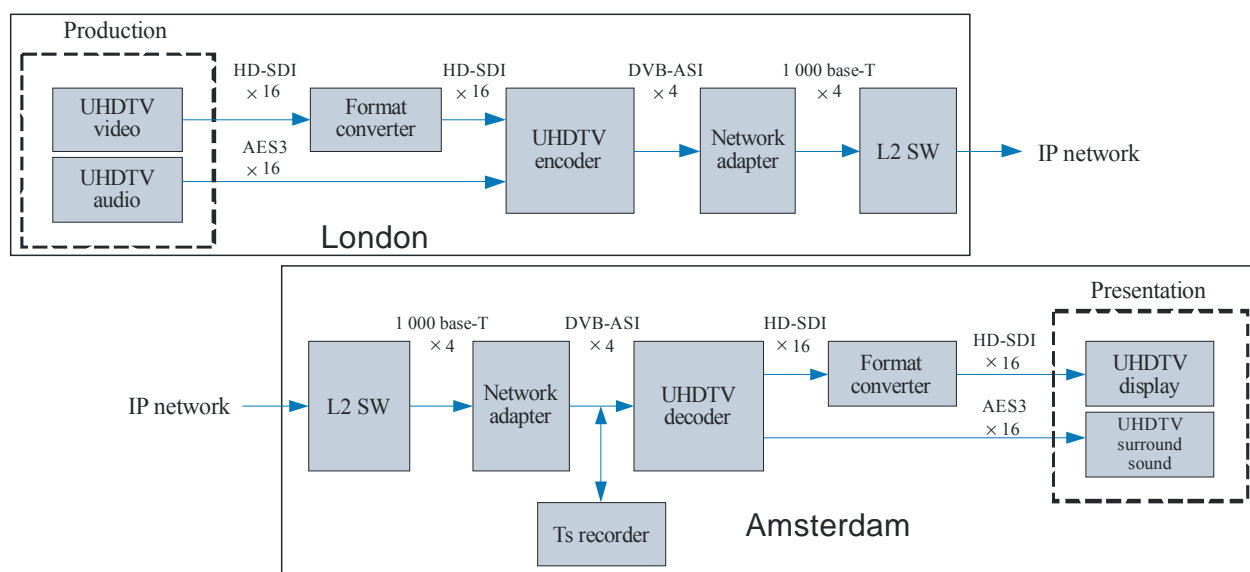
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6.2.3.2 Emulated live news reports demonstrated at IBC2008

Ultra high definition television (UHDTV) and 22.2 multichannel sound technologies were demonstrated at IBC2008 by the international collaborative group called the Broadcast Technology Futures group (BTF), which included international live contribution link over an ultra-broadband IP network. The outline is depicted in Fig. 19.

UHDTV live pictures and sound captured in central London were carried to Amsterdam over an ultra-broadband IP network. In order to demonstrate the live nature of the link, the scenario set up was to emulate live news reports from London to Amsterdam with two-way interaction between a reporter in London and a presenter in the theatre in Amsterdam.

FIGURE 19
UHDTV and 22.2 multichannel sound IP transmission system at IBC2008



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Sound acquisition system adopted in London was a microphone array with 15.2 system rather than a full-blown 22.2 system due to limitation on number of channels in mixing desk. This meant that there would be a middle layer containing eight of the ten specified 22.2 microphone complement, the top layer would be reduced to four microphones from nine, and the lower layer would have the full complement of five microphones, of which two were LFE channels. The microphone array is shown in Fig. 20. The total of 18 audio channels, including the 15.2 channels and one commentary channel, were sent to Amsterdam to be reproduced for the 22.2 multichannel system in the viewing theatre there. The 3D surround sound quality reproduced by 22.2 multichannel sound in Amsterdam was completely convincing; ambient sounds of London were reproduced effectively, even the sounds of airplanes and helicopters flying overhead sounded as if they were flying over the theatre.

6.3 Production of 10.2 multichannel sound

Programme material has been originally recorded for the 10.2 multichannel sound format, and some has been repurposed from other multichannel material. What is standardized is the playback platform including environment. No standardization of recording technique is required or desirable. A range of methods of recording were used, including adding microphones to more conventional 5.1-channel recording, layouts of microphones that mimic loudspeaker locations, and more complex pop style mixing wherein a large number of source microphone channels, up to 48 in several cases, are remixed to the 10.2 loudspeaker format.

There are more than 20 items of produced programme material. Since 10.2-channel sound is a playback platform, not a recording/playback system, a wide variety of methods of recording have been employed, from classic ones, to completely constructed spaces using advanced digital signal processing algorithms.

FIGURE 20
Microphone array



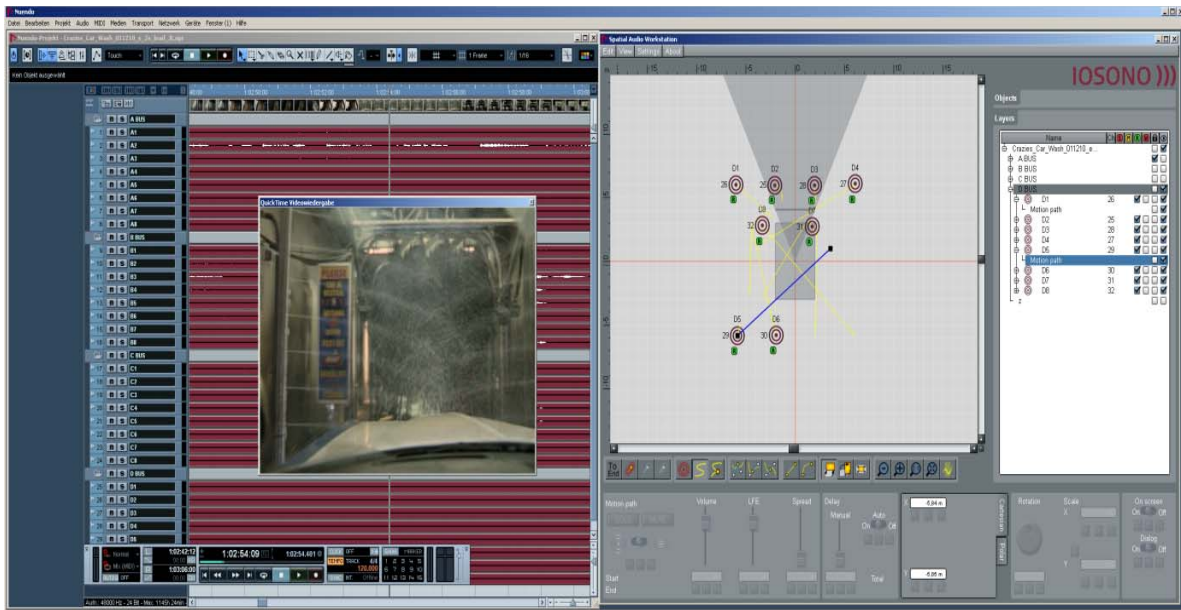
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6.4 Object-based post-production system

The creation of an object-based sound scene involves associating spatial information with the sound signals comprising the scene. The IOSONO Spatial Audio Workstation (SAW) is a tool for object-oriented production, editing and mastering of auditory scenes for reproduction in different environments. This plug-in for a digital audio workstation enables the direct monitoring of the object-based scene in all multichannel layouts. In combination with external rendering, the production can be performed directly in a flexible speaker layout or mixing stage. It is currently realized as a plug-in for the digital audio workstation Steinberg Nuendo. While Nuendo enables the editing and post production of audio streams, the SAW plug-in enables the sound engineer to create advanced sound source movements and complex audio scenes based on the audio material loaded into a Nuendo session (Fig. 21).

FIGURE 21

Spatial audio workstation used for a motion picture sound track production



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Using the SAW, sound objects are positioned on the scene like marking points on a map. In addition, the SAW allows the definition of motion trajectories for the sound objects. The mixer can assign a discrete position to each sound object for its x, y and z coordinates. For moving sound objects, the position information is accompanied by a timestamp (SMPTE time code). The user has full control over the sound objects and the motion lines. Moreover the plug-in offers a wide range of functions for sound objects and motion lines, e.g. move, rotate, scale and group. The SAW is equipped with a graphical user interface that allows the mixer to easily assign a discrete position to each sound object in the listeners' space. This gives the sound engineer an intuitive view compared to traditional channel-oriented loudspeaker panning techniques. With this tool, even live mixing is possible.

The output of the object-based production tool can be directly feed to any multichannel formats without additional processing. Sound engineers can switch the output format whenever they like without changing anything in the production. Combined with an external rendering, any reproduction system, including wave-field synthesis, can be used with the same content file.

FIGURE 22

Spatial audio workstation and WFS speaker array installed at Todd-AO Stage 2 used by Rick Kline



Report BS.2159-22

Several productions have been performed in Todd-AO (Fig. 22). Besides a number of trailers and demos, a complete motion picture sound-track has been produced using the spatial audio workstation (Fig. 22).

7 Quality performance of the multichannel sound systems

Subjective evaluations were conducted to assess the performance of three different multichannel audio systems: two-channel stereo, 5.1 channel sound, and 22.2 multichannel sound. The stimuli for the subjective evaluations were selected from the World Expo 2005 programmes. The sounds and video were screened at two locations: NHK Lab's UHDTV theatre (a 450-in. screen that is 14 m long, 15 m wide, and 10 m high) and NHK Lab's small post-production studio (a 50-in. screen that is 8 m long, 7.5 m wide, and 4 m high). The subjects were asked to report their impressions of the sound provided by the different sound systems when shown with different images on the screen. They were also asked to sit in different positions so that differences in impressions based on position with regard to the screen could be discussed.

The semantic difference evaluation method was used in this experiment. Subjects were asked to rate their impressions on a 7-point scale for the pairs of evaluation terms. Each pair contained two terms with opposite meanings, such as "dynamic" and "static." Subjects were asked to select a score from 3 to -3, in which 3 meant very dynamic, 2 meant fairly dynamic, 1 meant slightly dynamic, 0 meant neither dynamic nor static, -1 meant slightly static, -2 meant fairly static, and -3 meant very static. They rated each stimulus (segment of content) using the 24 evaluation pairs. There were 53 subjects (28 university students of music or audio engineering and 25 audio professionals) for this experiment.

Figures 23 and 24 show the total mean values of all the results from the 24 pairs of evaluation terms for each sound system in the large theatre and the small studio, respectively. Each mean value is marked with a 95% confidence interval for different terms and for different sound systems. The horizontal axis represents the scale of evaluation, and the vertical axis contains each pair of evaluation terms. Both figures show that the 22.2 multichannel sound system was rated significantly better (larger value) than the two-channel stereo system, for every evaluation term except “loud”. Figure 23 shows that, in the large theatre, the 22.2 multichannel sound system was rated significantly better than the 5.1 channel sound system for the terms “gaudy”, “distinct”, “wide in front and rear”, “wide in above and below”, “clear movement of sound”, “sound from every direction”, “rich reverberant” and “rich envelopment”. Figure 24 shows that, in the small studio, the 22.2 multichannel sound system was rated significantly better than the 5.1 channel sound system for every evaluation term except “loud”, “dynamic”, “gaudy” and “natural”. The results also show that the 5.1 channel sound system was rated significantly better than the two-channel stereo system for every term except “loud”. The results suggest that there was no difference in the loudness between each system. They also suggest that the 22.2 multichannel sound system provided a better 3D spatial sound quality than both the two-channel stereo and the 5.1 surround sound systems in both a large theatre and a small studio. Furthermore, the difference of rate between the 22.2 multichannel sound system and the 5.1 channel sound system or two-channel stereo system is basically bigger in a small studio than in a large theatre. Therefore, in a small studio, the 22.2 multichannel sound system may provide a better 3D spatial sound quality than both other sound systems than in a large theatre.

8 Relevant documents concerning the multichannel sound systems developed by organizations outside ITU

8.1 SMPTE

8.1.1 SMPTE 2036-2-2008, “Ultra High Definition Television – Audio characteristics and audio channel mapping for programme production”

SMPTE 2036 Ultra High Definition Television (UHDTV) suite of documents is in multiple parts:

- Part 1: Image parameter values for programme production.
- Part 2: Audio characteristics and audio channel mapping for programme production.

FIGURE 23

Results of subjective evaluation comparing three different sound systems with a large screen in a large theatre

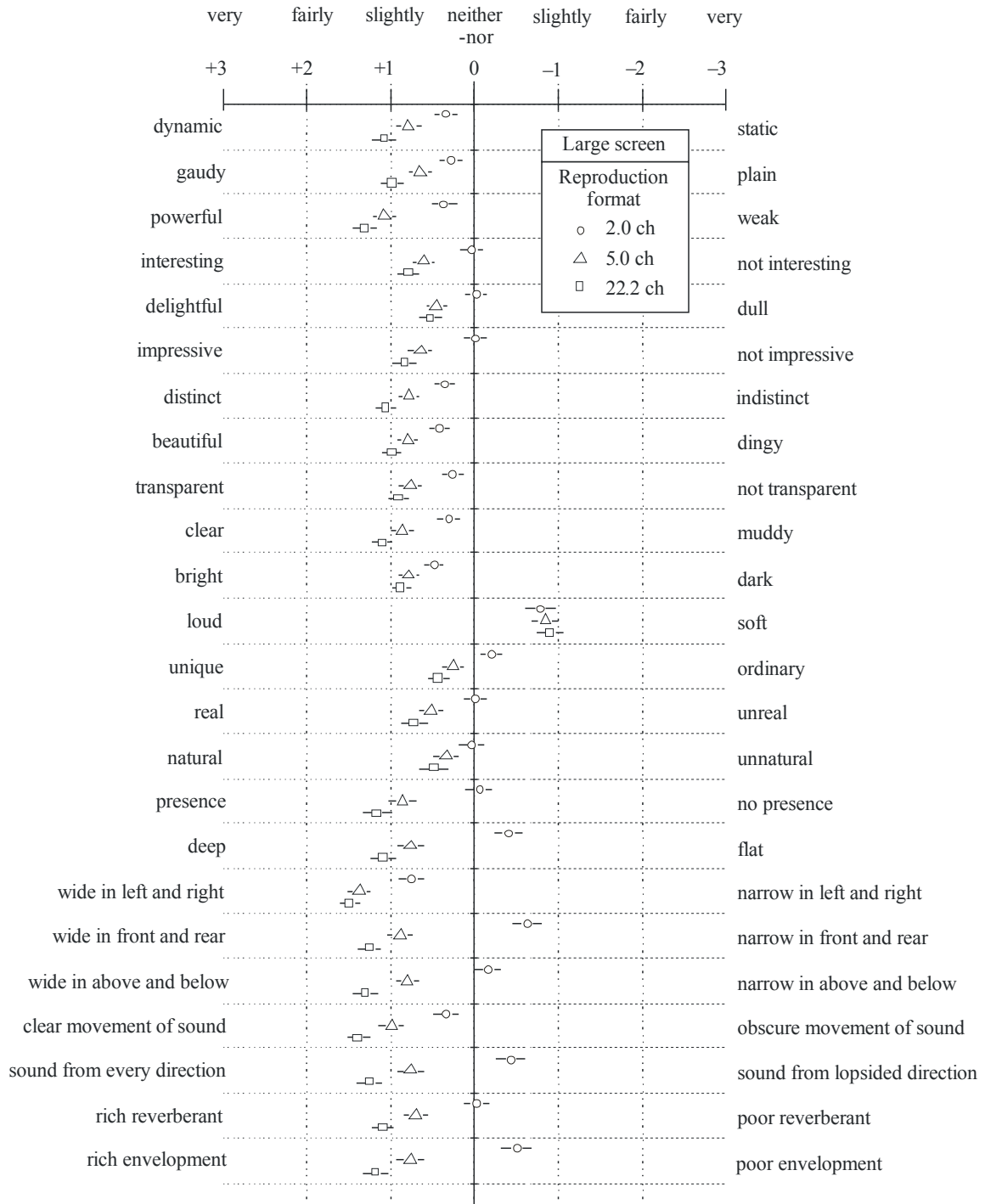
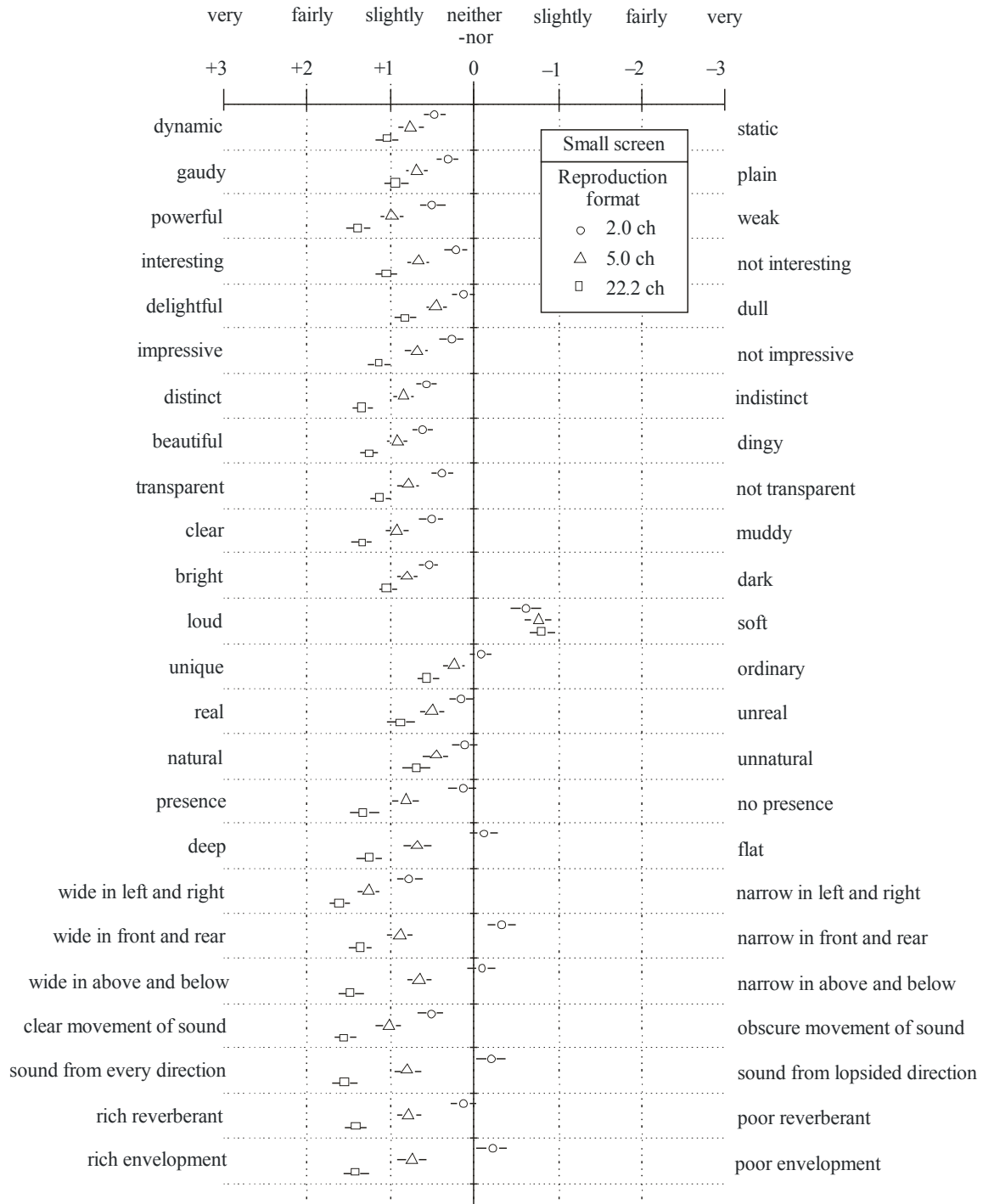


FIGURE 24

Results of subjective evaluation comparing three different sound systems with a small screen in a small studio



SMPTE Standard 2036-2-2008 is Part 2 of SMPTE 2036 and describes the audio characteristics and audio channel mapping for programme production. This document specifies the characteristics of digital audio for UHDTV programme production and distribution, and also defines the mapping and labelling of 22.2 multichannel audio for UHDTV programme production.

The audio specifications are as follows:

1. Digital signal characteristics
UHDTV audio shall support a channel count of 24 full-bandwidth channels.
NOTE 1 – The two LFE channels are transported as full-bandwidth channels.
2. Channel mapping and channel labelling of 22.2 multichannel audio

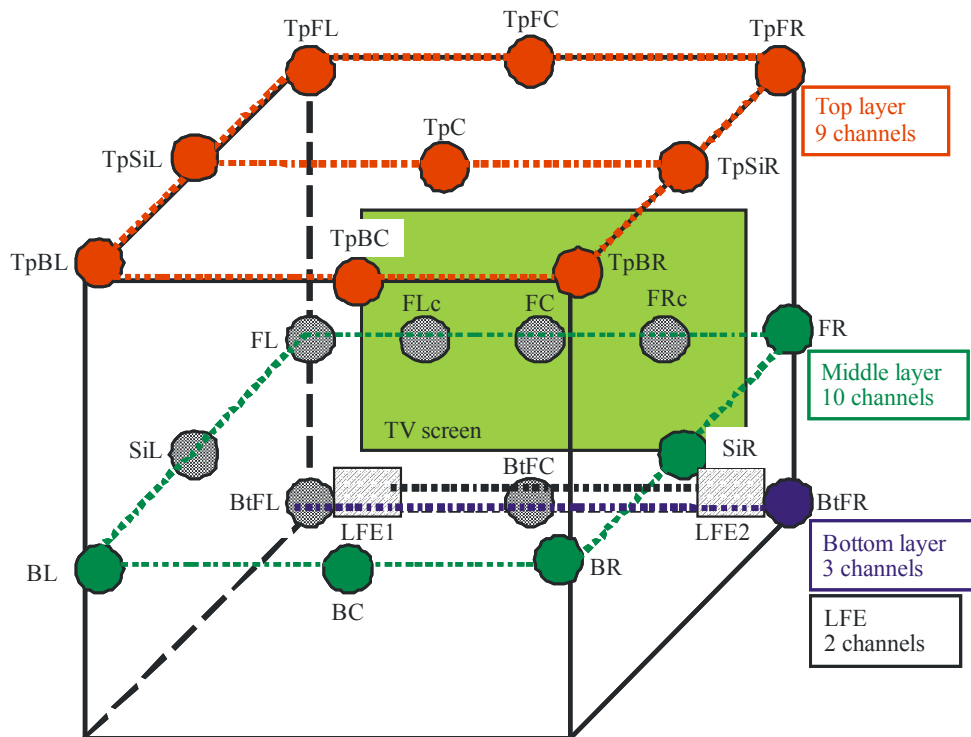
TABLE 2
Channel maps and labels of 22.2 multichannel audio

AES Pair No./Ch No.	Channel No.	Label	Name
1/1	1	FL	Front left
1/2	2	FR	Front right
2/1	3	FC	Front centre
2/2	4	LFE1	LFE-1
3/1	5	BL	Back left
3/2	6	BR	Back right
4/1	7	FLc	Front left centre
4/2	8	FRc	Front right centre
5/1	9	BC	Back centre
5/2	10	LFE2	LFE-2
6/1	11	SiL	Side left
6/2	12	SiR	Side right
7/1	13	TpFL	Top front left
7/2	14	TpFR	Top front right
8/1	15	TpFC	Top front centre
8/2	16	TpC	Top centre
9/1	17	TpBL	Top back left
9/2	18	TpBR	Top back right
10/1	19	TpSiL	Top side left
10/2	20	TpSiR	Top side right
11/1	21	TpBC	Top back centre
11/2	22	BtFC	Bottom front centre
12/1	23	BtFL	Bottom front left
12/2	24	BtFR	Bottom front right

3. Loudspeaker layout (informative)

Figure 25 illustrates the loudspeaker layout of a 22.2 multichannel sound system.

FIGURE 25



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8.2 IEC

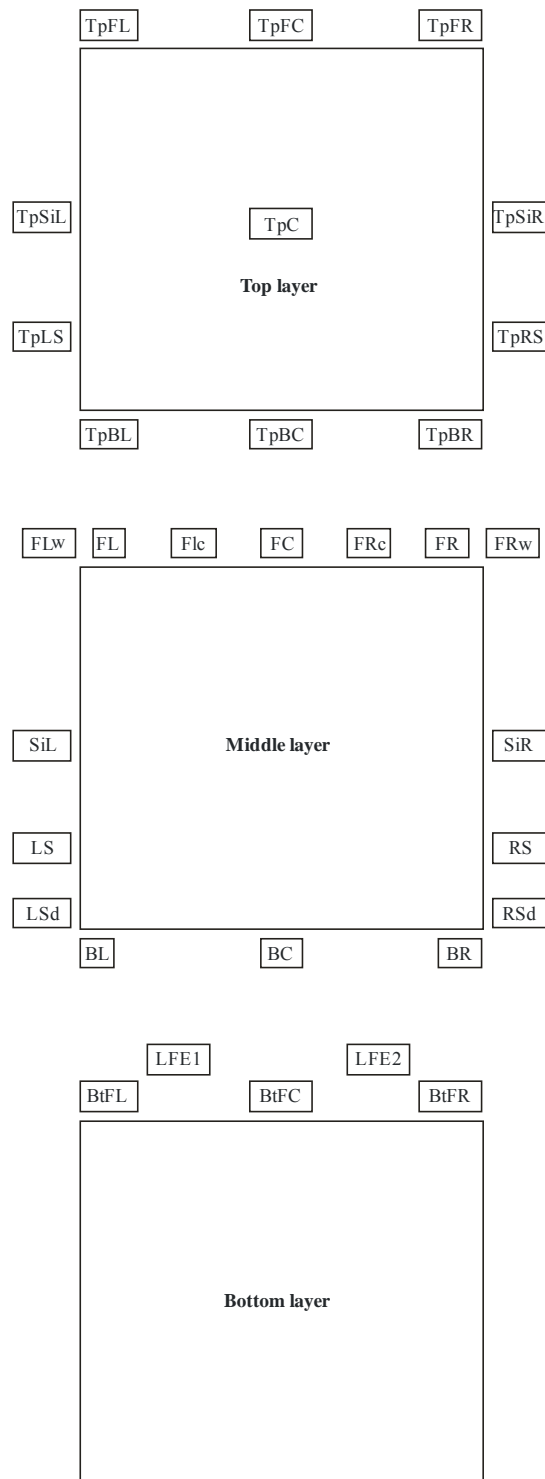
8.2.1 IEC 62574 “Audio, video and multimedia systems – General channel assignment of multichannel audio”

IEC 62574 “Audio, video and multimedia systems – General channel assignment of multichannel audio” specifies one general multichannel assignment. The general channel assignment as a channel mapping and labelling provides the unified usage of channel assignments for source devices, digital audio interfaces and sink devices. This standard excludes the specification of the exact position of each loudspeaker. It is aimed at consumer applications, but is not targeted for theatrical environments. Up to 32 labels for loudspeaker positions are specified, which can be used for all current multichannel formats.

Outline

IEC 62574 defines three layers, in each of which channels are assigned as depicted in Fig. 26. These are assigned channels and their labels, but they do not define exact loudspeaker positions.

FIGURE 26
Layers and channel assignment



Report BS.2159-26

Table 3 shows the channel maps and labels of the general channel assignment. Each channel label has an ID name as an abbreviation for the label.

TABLE 3
General channel assignment table

Channel number	Channel label ID name	Full name of ID
1/2	FL/FR	Front Left/Front Right
3/4	FC/LFE1	Front Centre/Low Frequency Effects-1
5/6	BL/BR	Back Left/Back Right
7/8	FLc/FRc	Front Left centre/Front Right centre
9/10	BC/LFE2	Back Centre/Low Frequency Effects-2
11/12	Sil/SiR	Side Left/Side Right
13/14	TpFL/TpFR	Top Front Left/Top Front Right
15/16	TpFC/TpC	Top Front Centre/Top Centre
17/18	TpBL/TpBR	Top Back Left/Top Back Right
19/20	TpSiL/TpSiR	Top Side Left/Top Side Right
21/22	TpBC/BtFC	Top Back Centre/Bottom Front Centre
23/24	BtFL/BtFR	Bottom Front Left/Bottom Front Right
25/26	FLw/FRw	Front Left wide/Front Right wide
27/28	LS/RS	Left Surround/Right Surround
29/30	LSd/RSd	Left Surround direct/Right Surround direct
31/32	TpLS/TpRS	Top Left Surround/Top Right Surround

8.3 MPEG (ISO/IEC JTC 1/SC 29/WG 11)

8.3.1 MPEG-2 AAC (ISO/IEC 13818-7)

MPEG-2 Advanced Audio Coding (AAC), which has been standardized by ISO and IEC, specifies low bit rate audio coding scheme which able to include up to 48 audio channels in one stream. The technical corrigendum ISO/IEC 13818-7:2006/COR 1 on Advance Audio Coding (AAC) has been approved to add the channel mapping for application specific channel configurations. The audio coding signal of 3D sound system including 22.2 multichannel audio can be transmitted by applying the application specific channel configuration.

Table 3 shows the example of an application specific channel alignment for a 22.2 channel configuration.

8.4 EBU

8.4.1 EBU TECH 3306-2007, “RF64: An extended File Format for Audio”

The RF64 file format fulfils the longer-term need for multichannel sound in broadcasting and archiving. An RF64 file has additions to the basic Microsoft RIFF/WAVE specification to allow for either, or both:

- more than 4 Gbyte file sizes when needed;
- a maximum of 18 surround channels, stereo down-mix channel and bitstream signals with non-PCM coded data. This specification is based on the *Microsoft Wave Format Extensible* for multichannel parameters.

TABLE 4

**Audio syntactic elements and channel alignment
for an application-specific 22.2 channel configuration**

Number of channels	Audio syntactic elements, listed in order received	Channel to speaker mapping
22+2	single_channel_element, channel_pair_element, channel_pair_element, channel_pair_element, channel_pair_element, single_channel_element, lfe_element, lfe_element, single_channel_element, channel_pair_element, channel_pair_element, single_channel_element, channel_pair_element, single_channel_element, single_channel_element, channel_pair_element	centre front speaker, left, right front centre speakers, left, right front speakers, left, right side speakers, left, right back speakers, back centre speaker, left front low frequency effects speaker, right front low frequency effects speaker, top centre front speaker, top left, right front speakers, top left, right side speakers, centre of the room ceiling speaker, top left, right back speakers, top centre back speaker, bottom centre front speaker, bottom left, right front speakers

The file format is designed to be a compatible extension to the Microsoft RIFF/WAVE format and to the BWF format and its supplements and additional chunks. It extends the maximum size capabilities of the RIFF/WAVE and BWF format allowing for multichannel sound in broadcasting and audio archiving.

RF64 can be used in the entire programme chain from capture to editing and play out and for short or long term archiving of multichannel files.

An RF64 file with a bext chunk becomes an MBWF (multichannel BWF) file.

The following are specifications about audio channels:

1. Definition of a new format, RF64.

The wave format extensible channel mask contains 18 “#define” settings specifying different outspeaker positions (or channel allocations).

Microsoft Wave Format Extensible Channel Mask

```

#define SPEAKER_FRONT_LEFT           0x00000001
#define SPEAKER_FRONT_RIGHT          0x00000002
#define SPEAKER_FRONT_CENTER          0x00000004
#define SPEAKER_LOW_FREQUENCY         0x00000008
#define SPEAKER_BACK_LEFT             0x00000010
#define SPEAKER_BACK_RIGHT            0x00000020
#define SPEAKER_FRONT_LEFT_OF_CENTER 0x00000040
#define SPEAKER_FRONT_RIGHT_OF_CENTER 0x00000080
#define SPEAKER_BACK_CENTER            0x00000100
#define SPEAKER_SIDE_LEFT              0x00000200
#define SPEAKER_SIDE_RIGHT            0x00000400

```



```

#define SPEAKER_TOP_CENTER           0x00000800
#define SPEAKER_TOP_FRONT_LEFT      0x00001000
#define SPEAKER_TOP_FRONT_CENTER    0x00002000
#define SPEAKER_TOP_FRONT_RIGHT     0x00004000
#define SPEAKER_TOP_BACK_LEFT       0x00008000
#define SPEAKER_TOP_BACK_CENTER     0x00010000
#define SPEAKER_TOP_BACK_RIGHT      0x00020000

```

8.5 Japan

8.5.1 Advancement of satellite digital broadcasting

The advancement of satellite digital broadcasting has been studied in Japan. Due to its utilization of state-of-the-art technologies, the advanced system is expected to enable the effective use of spectrum and the introduction of new services. The Ministry of Internal Affairs and Communications of Japan finally standardized the advancement of satellite digital broadcasting system in February 2009. The technical specifications were also standardized by the Association of Radio Industries and Businesses (ARIB) of Japan in July 2009.

The advanced system has a number of new features, including a transmission capacity increased by 30% or more due to the LDPC code and less roll-off factor, APSK modulations for a much higher capacity, higher resolution imagery and H.264|MPEG-4 AVC, three-dimensional surround sound, and the transport of IP packets targeted for multimedia storage.

The audio coding specifications are as follows.

TABLE 5

Audio coding for advancement of satellite digital broadcasting

Audio input	Maximum channels	22.2 channels per stream
	Sampling frequency	48 kHz
	Bit resolution	16, 20, or 24 bits
	Audio mode	Mono, stereo, multi-channel stereo (3.0, 4.0, 5.0, 5.1, 6.1, 7.1, 10.2, 22.2), dual- mono
Audio coding	Standard	MPEG-2 AAC, SBR (spectral band replication)
	Profile	LC