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## DTS-UHD Point Source Renderer

**EBU**  
OPERATING EUROVISION

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***ETSI***

650 Route des Lucioles  
F-06921 Sophia Antipolis Cedex - FRANCE

Tel.: +33 4 92 94 42 00 Fax: +33 4 93 65 47 16

Siret N° 348 623 562 00017 - NAF 742 C  
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Sous-Préfecture de Grasse (06) N° 7803/88

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## Foreword

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European Broadcasting Union  
CH-1218 GRAND SACONNEX (Geneva)  
Switzerland  
Tel: +41 22 717 21 11  
Fax: +41 22 717 24 81

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## Modal verbs terminology

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## 1 Scope

The present document defines an audio renderer associated with the DTS-UHD codec defined in ETSI TS 103 491 [1]. The inputs to the renderer are one or more sets of audio waveforms along with mixing instructions, and the output is a single set of waveforms mapped to a defined speaker configuration. Each set of waveforms may represent either audio channels or audio objects. The mixing instructions may vary with time and they come from metadata carried in the DTS-UHD bitstream and from other application interfaces described herein.

---

## 2 References

### 2.1 Normative References

References are either specific (identified by date of publication and/or edition number or version number) or non-specific. For specific references, only the cited version applies. For non-specific references, the latest version of the referenced document (including any amendments) applies.

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The following referenced documents are necessary for the application of the present document.

- [1] ETSI TS 103 491: "DTS-UHD Audio Format; Delivery of Channels, Objects and Ambisonic Sound Fields", version 1.1.1.
- [2] Pulkki, Ville: "Virtual Sound Source Positioning Using Vector Base Amplitude Panning", JAES Volume 45 Issue 6 pp. 456-466; June 1997.

### 2.2 Informative References

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The following referenced documents are not necessary for the application of the present document but they assist the user with regard to a particular subject area.

Not applicable.

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## 3 Definitions and abbreviations

### 3.1 Definitions

For the purposes of the present document, the following terms and definitions apply:

**audio object:** waveform or set of waveforms that have been assigned a location in space

**hull:** collection of points forming a surface mesh to be used as a basis for panning

**output feeds:** set of audio waveforms mapping one to one to the physical speakers

**panning:** distributing a point source location into a speaker layout

**physical speaker:** speaker that physically exists in the listening space

**point:** location on the surface of the unit sphere

**point source:** audio waveform with defined spatial coordinates within the listening room

NOTE: A point source is treated (mathematically) as a uniform spherical radiation pattern, though in reality most speakers have a pattern of less than 180 degrees on the face of the speaker. This reality is accounted for by limiting the panner range.

**render:** combine waveforms with their gain contributions to produce audio waveforms in a specified speaker configuration

**speaker:** transducer that converts an electrical signal into soundwaves

**speaker configuration:** defined list of physical speakers in the listening space

**triplet:** set of three points

**virtual speaker:** speaker that does not physically exist in the listening space

## 3.2 Abbreviations

For the purposes of the present document, the following abbreviations apply:

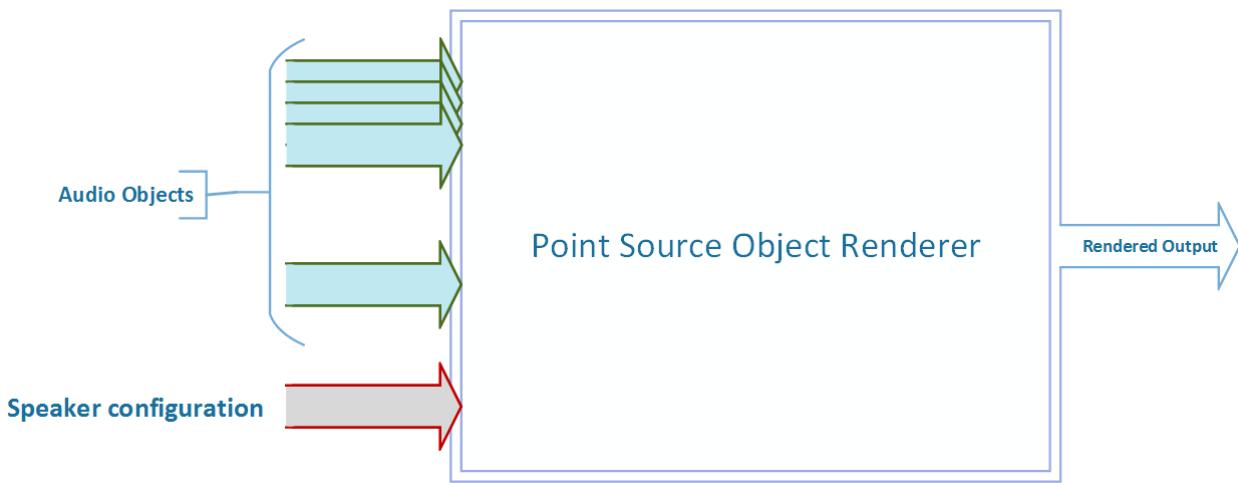
U	Union operator
3D	three dimensional
C	Centre speaker
Ch	Centre high speaker
Chr	Centre high rear speaker
Clf	Centre low front speaker
Cs	Centre surround speaker
dB	decibel
EMA	Exponential Moving Average
L	Left speaker
Lc	Left centre speaker
LFE	Low Frequency Effects
LFE1	Low Frequency Effects - 1 speaker
LFE2	Low Frequency Effects - 2 speaker
Lh	Left high speaker
Lhr	Left high rear speaker
Lhs	Left high side speaker
Llf	Left low front speaker
LR	Left Right
Ls	Left surround speaker
Lsr	Left surround rear speaker
Lss	Left side surround speaker
Ltf	Left top front speaker
Ltr	Left top rear speaker
Lw	Left wide speaker
N/A	Not Applicable
Oh	Overhead speaker
R	Right speaker
Rc	Right centre speaker
Rh	Right high speaker
Rhr	Right high rear speaker
Rhs	Right high side speaker
Rlf	Right low front speaker
Rs	Right surround speaker
Rsr	Right surround rear speaker
Rss	Right side surround speaker

Rtf	Right top front speaker
Rtr	Right top rear speaker
Rw	Right wide speaker

## 4 Renderer Metadata

### 4.1 Overview of the Renderer Inputs

The DTS-UHD point source object renderer is a system that computes the waveform associated with a specified speaker configuration, given a collection of audio objects as input, as shown in Figure 1. The renderer metadata assumes the model of a sphere with the listener's head at the centre (an ego-centric model). All speakers are treated as point sources and are assumed to be on the surface of a unit sphere, and thus have a radius of 1.



**Figure 1: Point Source Object Renderer System**

DTS-UHD audio objects consist of metadata plus one or more waveforms. The metadata always includes spatial location information, and may include additional parameters, such as gain information.

Audio objects can be manipulated by changing the gain and position metadata. For example, when the renderer metadata is coming from a DTS-UHD bitstream, the metadata can be updated as often as every 512 clock periods, where the clock shall be defined by `m_unClockRateInHz` in ETSI TS 103 491 [1]. If the audio within the DTS-UHD stream is carrying audio compressed using ACE, each ACE frame has a duration of 1 024 samples, therefore the mixing metadata can be updated twice per ACE frame. For audio with a sampling frequency of 48 KHz, this translates to a minimum update interval of 10,67 ms.

It is also possible for object metadata to be modified by user intervention. In this case, the bitstream metadata is being overridden until control is given back to the bitstream metadata.

The present document describes these in details to provide a comprehensive understanding of the DTS-UHD Point Source Object Renderer.

### 4.2 DTS-UHD Bitstream Metadata

DTS-UHD metadata is defined in ETSI TS 103 491 [1]. Since the renderer described in the present document is intended to be used in conjunction with the DTS-UHD bitstream, the DTS-UHD metadata parameters relating to the renderer are shown in Table 1.

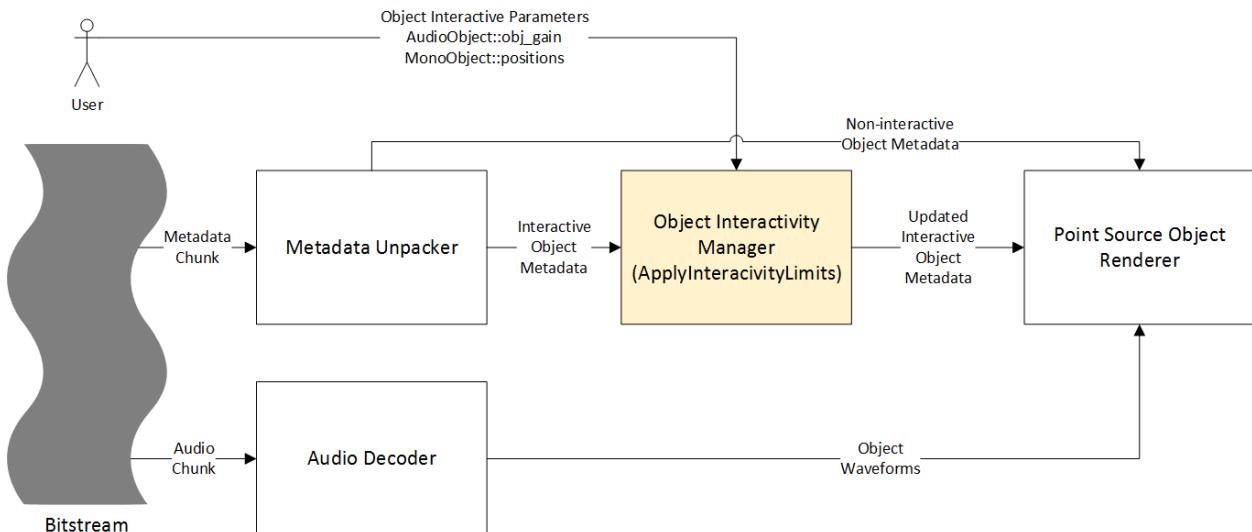
Note that Table 1 accommodates restrictions inherent in DTS-UHD metadata, but this does not imply such restrictions are imposed on the renderer capabilities.

**Table 1: DTS-UHD Metadata Parameters used for Rendering**

<b>Metadata Parameter</b>	<b>Reference in ETSI TS 103 491 [1]</b>	<b>Description</b>	<b>Object Renderer Parameter</b>
m_unObjectID	7.8.7	Identifier to access and modify an object state.	AudioObject::object_id
m_bMonoObjWithMultipleSourcesFlag (or bObjMS for short)	7.8.11.2	Indicates whether the AudioObject with one MonoObject is associated with multiple 3D points.	used with m_ucNum3DSourcesInObj (below)
m_ucNumWaveFormsInObj	7.8.11.17	Specifies the number of waveforms (MonoObjects) in the AudioObject.	AudioObject:: mono_objects[].size()
m_bPerSamplPeriodObjMDUpdFlag	7.8.11.22.2	Flag indicates that there are metadata updates per processing block within a larger audio block.	N/A
m_ppObjGain	7.8.11.22.3	Specifies the AudioObject gain.	AudioObject:: object_gain
m_ucNum3DSourcesInObj	7.8.11.22.4	Specifies the total number of point source locations in the AudioObject. Note that DTS-UHD metadata restricts an AudioObject to have either multiple MonoObjects each with a single point source location, or a single MonoObject with multiple point source locations.	if (bObjMS) { MonoObject:: positions[].size() AudioObject:: mono_objects[].size() = 1 } else { AudioObject:: mono_objects[].size() MonoObject:: positions[].size() = 1 }
unsrc_index	7.8.11.22.5	Identifies the source index within the AudioObject.	N/A
m_rPerObjExpWinLambda	7.8.11.23	Specifies the smoothing factor for the AudioObject.	AudioObject:: obj_lambda
m_3DSrcRadius	7.8.11.28.2	Specifies radius of the point source location.	AudioObject:: mono_objects[]. positions[].radius
m_3DSrcAzimuth	7.8.11.28.3	Specifies azimuth of the point source location. Range is [-180°, 180°].	AudioObject:: mono_objects[]. positions[].azimuth
m_3DSrcElevation	7.8.11.28.4	Specifies elevation of the point source location. Range is [-90°, 90°].	AudioObject:: mono_objects[]. positions[].elevation

### 4.3 DTS-UHD Object Interactivity Manager

Renderer APIs may allow the object metadata from the DTS-UHD bitstream to be overridden by the user during playback. Metadata in the bitstream may also be set to limit or disable a user interaction. The object interactivity manager enforces these rules and applies any user changes to the metadata before calling the renderer. Figure 2 shows that the object interactivity manager sits just before the renderer, where it handles the user input and the limit rules specified by the bitstream creator.



**Figure 2: Object Interactivity Manager**

Table 2 shows the interactive parameters specified in the DTS-UHD bitstream.

**Table 2: Object Interactive Parameters Specified in the Bitstream**

Input	Reference in ETSI TS 103 491 [1]	Description
m_bObjInteractiveFlag	7.8.11.12.1	Determines whether parameters can interactively change.
m_bObjInterLimitsFlag	7.8.11.12.2	Determines whether parameters have range limits constraints.
m_unMaxInterObjGainBoostdB	7.8.11.12.3	Specifies the maximum gain boost permitted.
m_unMaxInterObjGainAttendB	7.8.11.12.4	Specifies the maximum gain attenuation permitted.
m_unObjInterPosMaxDeltaAzim	7.8.11.12.5	Specifies the maximum allowed change in azimuth.
m_unObjInterPosMaxDeltaElev	7.8.11.12.6	Specifies the maximum allowed change in elevation.

The following pseudocode defines the function of an object interactivity manager.

```

ApplyInteractivityRules()
{
    // For each object in the bitstream
    for ( bitstream_obj : bitstream_obj_list )
    {
        // Only if this flag is true, the object is allowed to be interactive
        if ( bitstream_obj.m_bObjInteractiveFlag )
        {
            if ( bitstream_obj.m_bObjInterLimitsFlag )
            { // If this flag is set there are limits to the interactive parameters

                gain_max = bitstream_obj.m_ppObjGain +
                           bitstream_obj.m_unMaxInterObjGainBoostdB;
                gain_min = bitstream_obj.m_ppObjGain -
                           bitstream_obj.m_unMaxInterObjGainAttendB;

                user_interactive_obj_gain =
                    clip( user_interactive_obj_gain, gain_min, gain_max );
            }

            // Fetch PointSourceObjectRenderer's AudioObject to be updated
            AO = GetAudioObject( bitstream_obj.m_unObjectID );

            if ( has_user_specified_obj_gain ) {
                // Update gain
                AO.obj_gain = user_interactive_obj_gain;
            }

            // for each MonoObject
            for ( MO : AO.mono_objects )
            {

```

```
// for each point in the MonoObject
for ( p : MO.positions )
{
    if ( bitstream_obj.m_bObjInterLimitsFlag )
    { // If this flag is set there are limits to the
        // interactive params

        // Limit azimuth
        az_min = p.azimuth -
            bitstream_obj.m_unObjInterPosMaxDeltaAzim;
        az_max = p.azimuth +
            bitstream_obj.m_unObjInterPosMaxDeltaAzim;
        user_interactive_obj_position.azimuth =
            clip( user_interactive_obj_position.azimuth, az_min, az_max );

        // Limit Elevation
        el_min = p.elevation -
            bitstream_obj.m_unObjInterPosMaxDeltaElev;
        el_max = p.elevation +
            bitstream_obj.m_unObjInterPosMaxDeltaElev;
        user_interactive_obj_position.elevation =
            clip( user_interactive_obj_position.elevation, el_min, el_max );
    }

    if (has_user_specified_obj_position)
    {
        // update position
        p = user_interactive_obj_position;
    }
}
}
```

---

## 5 Variables and Class Types

### 5.1 Global Variables

The following data types and variables are used throughout the present document:

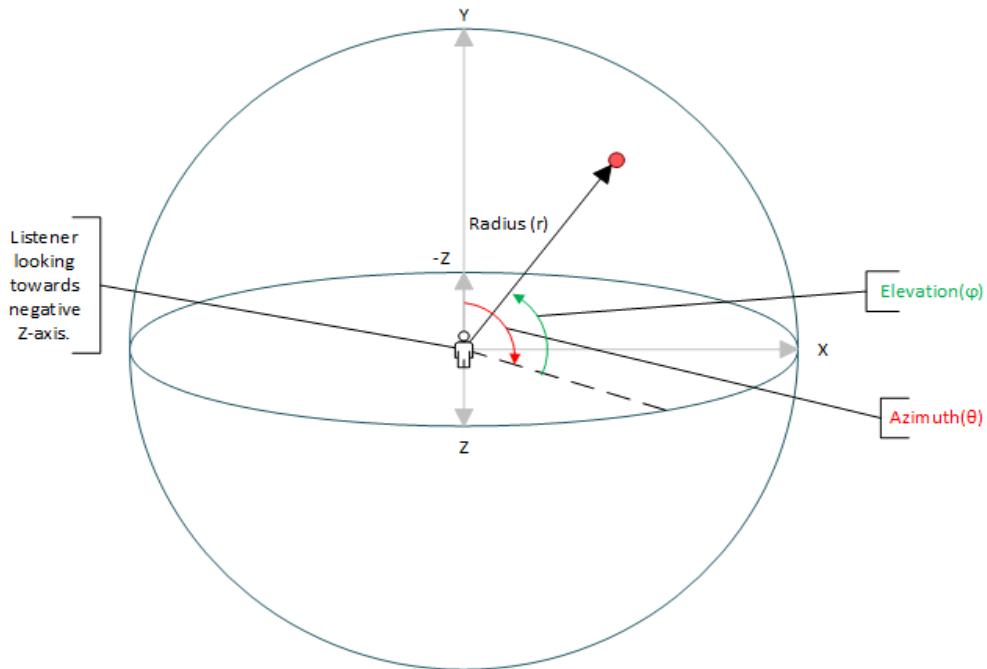
- **AO:** An object of type AudioObject.
- **H:** An object of type Hull.
- **MO:** An object of type MonoObject.
- **PSOR:** An object of type PointSourceObjectRenderer.
- **SC:** An object of type SpeakerConfiguration. A speaker configuration consists of a list of speaker locations that physically exist in the listening space.
- **V:** List of all virtual speakers or  $V = V_e \cup V_u \cup V_l$ .
- **V<sub>e</sub>** or v\_e: List of virtual speakers in the equatorial plane.
- **V<sub>l</sub>** or v\_l: List of virtual speakers in the lower hemisphere.
- **V<sub>u</sub>** or v\_u: List of virtual speakers in the upper hemisphere.
- **VBP:** An object of type VectorBasedPanner.

## 5.2 Renderer Object Classes

### 5.2.1 Point

A point is a location of any entity (for example, speaker or audio object) in the listening space. It is specified by polar coordinates - azimuth ( $\theta$ ) elevation ( $\phi$ ) and radius ( $r$ ). For the purposes of the present document, only the points on the unit sphere are needed, which means the radius shall be equal to 1. The listener is at the origin ( $\theta = 0, \phi = 0, r = 0$ ), facing the location ( $\theta = 0, \phi = 0, r = 1$ ).

Figure 3 shows the coordinate system of the renderer. Note that the equatorial plane is represented with  $\phi = 0$ . The upper hemisphere is specified as  $\phi > 0$ , and lower hemisphere is specified as  $\phi < 0$ .



**Figure 3: Point Source Object Renderer Coordinate System**

The following pseudocode defines a point class. For convenience in determining the distance between points, a conversion to Cartesian coordinates is provided below:

```
class Point {
    float azimuth;
    float elevation;
    float radius;

    Cartesian();
};

// Converting to Cartesian coordinate system.
Point::Cartesian()
{
    float x = radius * sin(azimuth) * cos(elevation);
    float y = radius * sin(elevation);
    float z = radius * -cos(azimuth) * cos(elevation);
    return (x,y,z);
}
```

## 5.2.2 Edge

An Edge is a line segment connecting two points. For the purposes of the present document, the edge structure shall be a set of two Points.

```
struct Edge
{
    // Constructor
    Edge(Point p1, Point p2) : p1(p1), p2(p2) {}

    // The two points representing the edge.
    Point p1, p2;

    // Equals operator
    bool operator==(Edge other)
    {
        return ((p1 == other.p1 && p2 == other.p2) || (p1 == other.p2 && p2 == other.p1));
    }
};
```

## 5.2.3 SpeakerConfiguration

A speaker configuration lists the location of physical speakers in the listening space. This shall be established when initializing the renderer. Speaker configuration is represented as follows:

```
class SpeakerConfiguration
{
    Point speaker_locations[];
    uint32 GetChannelMask();
};

SpeakerConfiguration::GetChannelMask()
{
    uint32 channel_mask = 0;
    for ( sl : speaker_locations ) {

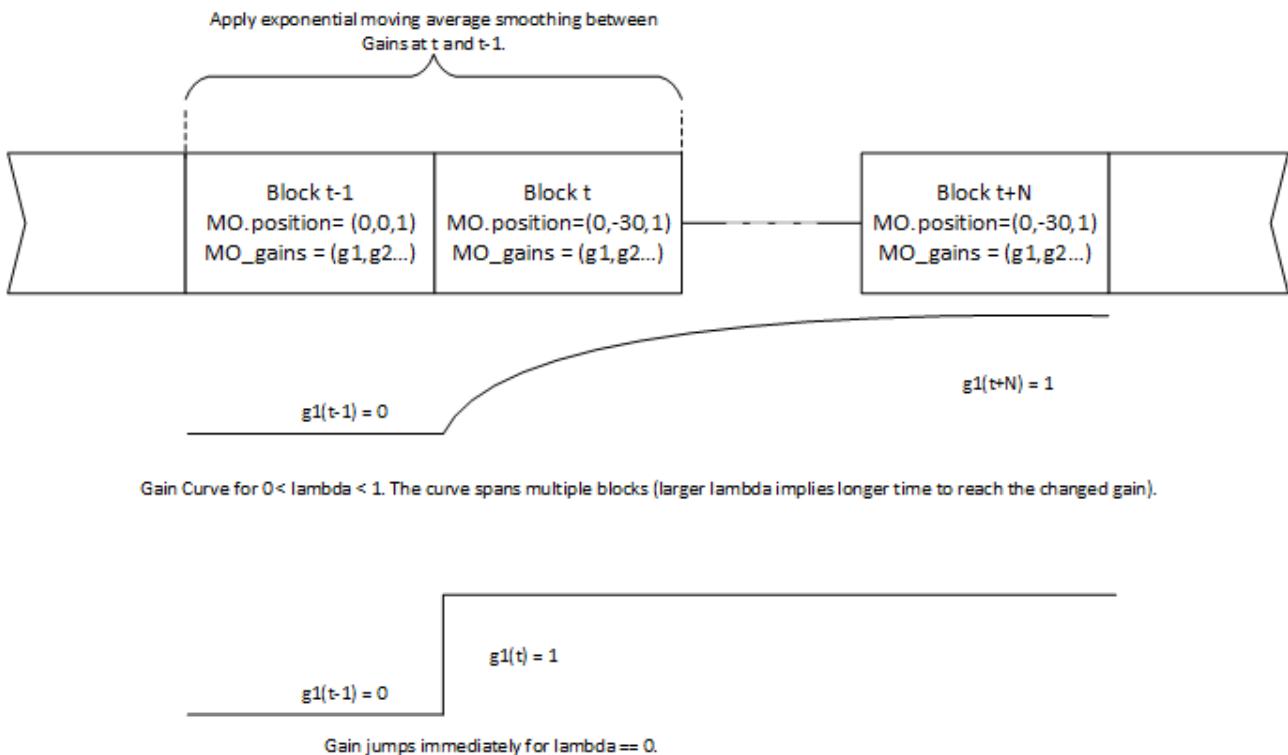
        // GetChannelMask(sl) fetches channel mask for predefined set of locations in Table 3.
        // If the location is not specified, then returns 0.

        uint32 location_ch_mask = GetChannelMask(sl);

        if (location_ch_mask)
            channel_mask |= location_ch_mask;
        else {
            channel_mask = 0;
            break;
        }
    }
    return channel_mask;
}
```

## 5.2.4 EMASmooth

EMASmooth is a class supporting an Exponential Moving Average (EMA) smoothing function. The renderer processes each audio object in blocks. A block contains a set of audio samples with associated metadata. The size of the block, defined as the number of audio samples in the block, is specified at the rendering stage. The block size is dynamic, adapting to the frequency of metadata updates. The audio object properties, such as object gain or mono object locations, can change from block to block. This dynamic change results in a change of gain coefficients between blocks. At these transitions, the renderer uses an exponential moving average smoothing algorithm, defined as Smooth(), to smooth the gains. In Figure 4 the position of MonoObject changes between the block at time  $t-1$  and the block at time  $t$ , affecting the gain contributions. For example, the gain contribution of MonoObject to the first speaker in the output may change from  $g1(t-1) = 0$  to  $g1(t) = 1$ . The EMASmooth curve for this transition is shown in Figure 4. The curve may span multiple blocks to settle on the new gain depending on lambda and block\_size.



**Figure 4: EMASmooth Sample Application**

The pseudocode below provides the implementation of the exponential moving average smoother. The smoothing factor, lambda, may change at every block.

The previous gain value (prev\_gain) is initialized to 0 when the object is created. This results in a ramp up to the new value when calculated. The user can choose not to ramp up by setting lambda to 0.

```
class EMASmooth
{
    float lambda;
    float prev_gain;
    Smooth(float new_gain, uint block_size);
};

EMASmooth::Smooth(float new_gain, uint block_size)
{
    float smoothed_gain_over_a_block[block_size];

    for ( uint t : 0 to block_size )
    {
        smoothed_gain_over_a_block[t] = new_gain * (1-lambda) + prev_gain * lambda;
        prev_gain = smoothed_gain_over_a_block[t];
    }
    return smoothed_gain_over_a_block;
}
```

## 5.2.5 MonoObject

A mono object consists of a single waveform with a list of associated points. A mono object can have more than one point source location. The following structure defines the mono object class.

```
class MonoObject
{
    // List of point sources locations.
    Point positions[];

    // Single waveform - array of audio samples for the block (an audio frame)
    float waveform[];
    // Smoother list - a persistent parameter that exist through the lifetime of the MonoObject.
    // It is of length equal to the SpeakerConfiguration::speaker_locations.size().
    EMASmooth smoother_list[];
```

```
};
```

## 5.2.6 AudioObject

An audio object is defined as a set of mono objects. An audio object consists of object properties such as, `object_id`, `object_gain`, `object_lambda` (smoothing factor) along with a list of `MonoObjects`. The following structure defines the `AudioObject` class:

```
class AudioObject
{
    MonoObject mono_objects[];
    float obj_lambda;
    float obj_gain;
    uint object_id;
};
```

## 5.2.7 Vector

The definition of the `Vector` class is provided below without an implementation. It supports basic vector operations, such as dot product and cross products.

```
// A Vector class defined without implementation.
class Vector
{
    Dot( Vector other ); // Computes dot product
    Cross( Vector other ); // Computes cross product
    int data[]; // internal data accessed through operator[]
};
```

## 5.2.8 Matrix

The definition of the `Matrix` class is provided below without an implementation. It supports basic matrix multiplication and inverse functions. The `Matrix` class shall have fixed dimensions of three rows by three columns.

```
// A matrix class defined without implementation.
class Matrix
{
    Column(col); // Peek the column col
    Row(row); // Peek at the row
    IsInvertible(); // True if the matrix is invertible.
    Invert(); // Returns inverse of a matrix
    Operator*( Matrix other ); // Multiply with matrix 'other'.
    int data[][][]; // internal data accessed through operator[]
};
```

## 5.2.9 Triplet

A triplet is a spherical triangle formed on the unit-sphere surface specified by a set of three non-collinear points. Triplets are used within `Hull` (clause 5.2.10) to represent its surface mesh. `Triplet` is defined as follows:

```
class Triplet
{
    Point p1;
    Point p2;
    Point p3;

    // return Matrix object
    GetTripletMatrix();

    // rearrange points so that triangle formed by them are positively oriented
    ConvertToPositiveOrientation();

    // Triplet in equator or upper hemisphere
    FullyInUpperHemisphere();

    // Triplet in equator or lower hemisphere
    FullyInLowerHemisphere();

    // Triplet in equator
    FullyInEquatorialPlane();
```

```
};
```

### **Triplet::GetTripletMatrix**

```
Triplet::GetTripletMatrix()
{
    Matrix M;
    M.Column(0) = p1.Cartesian();
    M.Column(1) = p2.Cartesian();
    M.Column(2) = p3.Cartesian();
    return M;
}
```

### **Triplet::ConvertToPositiveOrientation**

```
// Convert triplets into positive-oriented convex polygon.
Triplet::ConvertToPositiveOrientation()
{
    Vector v1 = p1.Cartesian();
    Vector v2 = p2.Cartesian();
    Vector v3 = p3.Cartesian();

    Vector V[] = {v1, v2, v3, v1}; // cyclic

    // Find the sign of first Vector
    Vector n = V[0].Cross(V[1]);
    dots = [];
    for ( i : range(V.size()) )
        dots[i] = V[i].Dot(n);
    sign = GetSign(dots);

    // Match the sign with other Vectors
    for ( i : range(1, V.size()-1) )
    {
        n = V[i].cross(V[i+1]);
        for ( i : range(V.size()) )
            dots[i] = V[i].Dot(n);

        if ( sign != GetSign(dots) )
            sign = 0;
    }

    // negative oriented reverse the order of p1,p2,p3
    if (sign == -1)
        Swap(p1, p3);
}
```

```
// Return +1 if all positive
// Return -1 if all negative
// Return 0 if mixed or all zero
GetSign(float arr[])
{
    all_pos = true;
    all_neg = true;
    for ( a : arr ) {
        if ( a < 0 )
            all_pos = false;
        if ( a > 0 )
            all_neg = false;
    }
    sign = 0;
    if ( all_pos && !all_neg )
        sign = 1;
    if ( !all_pos && all_neg )
        sign = -1;
    return sign;
};
```

### **Triplet::FullyInUpperHemisphere**

For the purposes of this function, the upper hemisphere includes the equator.

```
Triplet::FullyInUpperHemisphere()
{
    return (p1.elevation >= 0 && p2.elevation >= 0 && p3.elevation >= 0);
```

### **Triplet::FullyInLowerHemisphere**

For the purposes of this function, the lower hemisphere includes the equator.

```

Triplet::FullyInLowerHemisphere()
{
    return (p1.elevation <= 0 && p2.elevation <= 0 && p3.elevation <= 0);
}

Triplet::FullyInEquatorialPlane

Triplet::FullyInEquatorialPlane()
{
    return (p1.elevation == 0 && p2.elevation == 0 && p3.elevation == 0);
}

```

## 5.2.10 Hull

A hull is a set of points formed by combining the physical loud speaker locations and the virtual speaker locations. These points are used to form a surface mesh by grouping them into triplets. The structure below defines the `Hull` class.

```

class Hull
{
    Triplet T[]; // list of triplets
    Point hull_points[]; // list of points that form the hull
    Float epsilon; // epsilon is used to handle floating point errors

    // Initialize the hull
    Init(Point hull_points[])

    // Pan the point p into the hull
    Pan(Point P);
};

```

## 5.2.11 Virtual Speakers

A virtual speaker for a given speaker configuration is specified by the position on the unit sphere and the fold-down contributions to the specified physical speaker configuration. Virtual speakers enable the vector based panner to create a complete three-dimensional hull. The vector based panner is supported by two virtual speaker classes.

The first class is `PredefinedVirtualSpeakers`, which supports virtual speaker generation for a specific set of output speaker configurations. The structure below defines the class `PredefinedVirtualSpeakers`. The definition and use of the predefined virtual speakers is further defined in clause 6.3.2.

```

class PredefinedVirtualSpeakers
{
    // Check if the speaker configuration is supported
    IsSupported(SpeakerConfiguration SC);

    // Find virtual speakers and fold-down coefficients
    GetVirtualSpeakersAndFolddownCoeffs(SpeakerConfiguration SC);
}

The other class of virtual speaker is AutoVirtualSpeaker. When the speaker configuration is not supported by predefined virtual speakers, this second algorithm is used. This algorithm can support arbitrary speaker configurations. The structure below defines the class AutoVirtualSpeakers. The algorithm and use of auto virtual speakers is further defined in clause 6.3.3.

class AutoVirtualSpeakers
{
    // Find virtual speakers and fold-down coefficients
    GetVirtualSpeakersAndFolddownCoeffs(SpeakerConfiguration SC);
}

```

## 5.2.12 VectorBasedPanner

The renderer uses vector based panning to calculate gain contributions of every point source location to the given output speaker configuration. The algorithm for the vector based panner is further defined in clause 6.5.2.

The following pseudocode defines the `VectorBasedPanner` class.

```

class VectorBasedPanner
{
    SpeakerConfiguration SC;
    Point V[]; // virtual points
    float FD[][]; // fold-down coefficients for each virtual point
    Hull H; // Hull used for panning
}

```

```

// Initialize the panner
Init(SpeakerConfiguration SC);

// Pan the point p
Pan (Point p);
};

```

### 5.2.13 PointSourceObjectRenderer

The class `PointSourceObjectRenderer` shows the interface to the renderer. There are two public APIs:

- `Init()` - renderer initialization
- `RenderAllObjects()` - rendering objects

The calls to the renderer are defined below:

```

class PointSourceObjectRenderer
{
public:
    // Initialize the renderer.
    Init(SpeakerConfiguration sc);

    // Render the objects in AO_list.
    RenderAllObjects(AudioObject AO_list[], uint block_size );

private:
    // Render a single AudioObject.
    RenderAudioObject(AudioObject AO, uint block_size );

    // Render a MonoObject.
    RenderMonoObject(
        MonoObject MO,
        uint block_size,
        float obj_gain,
        float obj_lambda );

    // Speaker configuration this instance of object renderer is initialized with.
    SpeakerConfiguration sc;

    // Vector based panner for a given speaker configuration.
    VectorBasedPanner vbp;

    // The output waveforms in the specified speaker configuration.
    float out_waveforms[][][];
};


```

## 5.3 Other Helper Functions

### 5.3.1 Join

`Join()` combines multiple collections of objects into one container. A simple pseudocode for `Join()` is provided below:

```

// Join lists in a two dimensional array
// or arrays specified in comma separated list.
Join( collection[][] )
{
    res = [];
    for ( i : collection.size() )
    {
        for ( val : collection[i].size() )
            res += val;
    }
    return res;
}

```

### 5.3.2 MapAzimuthTo0\_360

`MapAzimuthTo0_360()` converts azimuth to all positive angles. Angles in the range  $[-180^\circ : 0^\circ]$  are mapped to  $[180^\circ : 360^\circ]$ .

```
MapAzimuthTo0_360(Point points[])
{
    modified_points = []
    for (p : points)
    {
        new_az = (p.azimuth - floor(p.azimuth / 360) * 360);
        modified_points += { new_az, p.elevation, p.radius };
    }
    return modified_points;
}
```

### 5.3.3 GetEquatorialPoints

`GetEquatorialPoints()` returns the points on the equator.

```
GetEquatorialPoints(Points points[])
{
    equatorial_points = [];
    for (p : points)
    {
        if (p.elevation == 0)
            equatorial_points += p;
    }
    return equatorial_points;
}
```

### 5.3.4 NumUpperHemispherePoints

`NumUpperHemispherePoints()` returns the number of points in the upper hemisphere. For the purposes of this function, this excludes points on the equator.

```
NumUpperHemispherePoints(Point points[])
{
    count = 0;
    for (p : points)
    {
        if (p.elevation > 0)
            ++count;
    }
    return count;
}
```

### 5.3.5 NumLowerHemispherePoints

`NumLowerHemispherePoints()` returns the number of points in the lower hemisphere. For the purposes of this function, this excludes points on the equator.

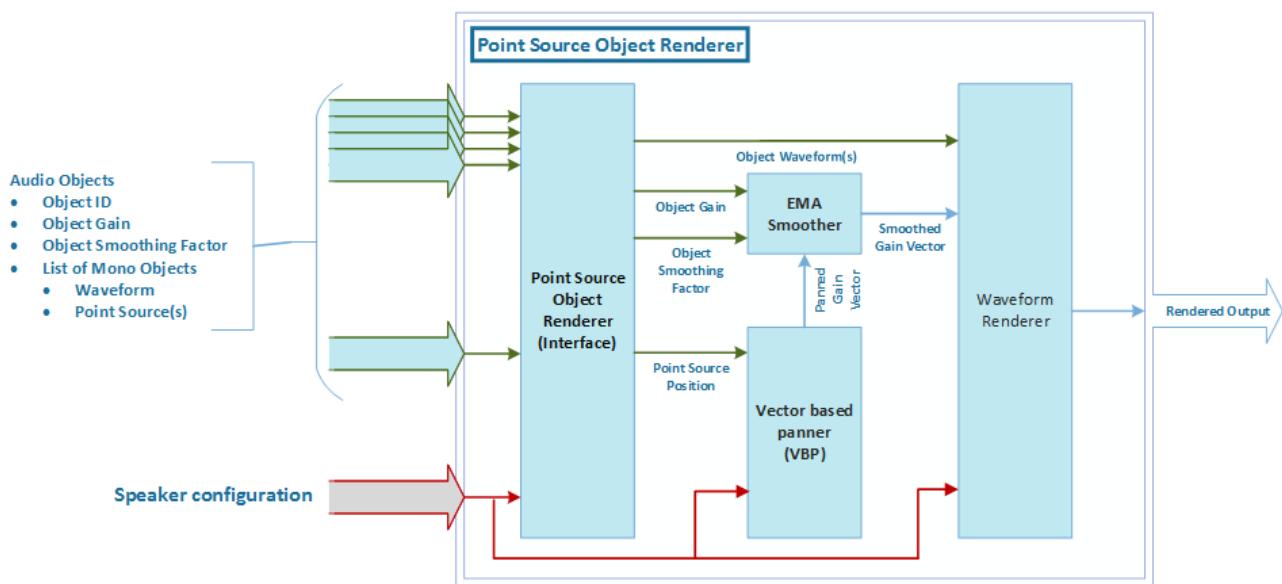
```
NumLowerHemispherePoints(Point points[])
{
    count = 0;
    for (p : points)
    {
        if (p.elevation < 0)
            ++count;
    }
    return count;
}
```

## 6 Point Source Object Renderer

### 6.1 Operation of the Renderer

The rendering system shall be initialized with an output SpeakerConfiguration before it is used. After initialization, a list of audio objects is rendered one block at a time. The underlying system that enables initialization and rendering is shown in Figure 5. The interface to the renderer consists of two APIs:

- `Init()`
  - Initializes VectorBasedPanner
- `RenderAllObjects()`
  - Pans using vector based panner
  - Applies exponential smoothing
  - Renders the objects into the output feeds



**Figure 5: Point Source Object Renderer Block Diagram**

A higher-level usage of `PointSourceObjectRenderer` is shown in the pseudocode below. Here the renderer is initialized with a speaker configuration object followed by the actual rendering.

```

// Declare PointSourceObjectRenderer
PointSourceObjectRenderer PSOR;

// Define SpeakerConfiguration
SpeakerConfiguration SC = {
    speaker_location1,
    speaker_location2,
    ...
    speaker_locationN
};

// Initialize renderer
PSOR.Init(SC);

// Run renderer for all blocks
for ( audio_objects[] : audio_object_blocks )
{
    // Render all objects
    PSOR.RenderAllObjects(audio_objects);
}

```

## 6.2 Renderer Initialization

The initialization of the renderer involves identifying virtual speakers and setting up the hull for panning, both of which are handled during initialization of `VectorBasedPanner`. The `PointSourceObjectRenderer::Init()` is initializing `VectorBasedPanner` as shown in the pseudocode below:

```
PointSourceObjectRenderer::Init(SpeakerConfiguration SC)
{
    this->SC = SC;
    VBP.Init(SC);
}
```

Initialization of `VectorBasedPanner` takes the speaker configuration and determines the number and location of any required virtual speakers and their fold-down coefficients. The virtual speakers are then used to initialize the convex hull, refer to the pseudocode below:

```
VectorBasedPanner::Init(SpeakerConfiguration SC)
{
    this->SC = SC;

    if (PredefinedVirtualSpeakers::IsSupported(SC))
        V, FD = PredefinedVirtualSpeakers::GetVirtualSpeakersAndFolddownCoeffs(SC);
    else
        V, FD = AutoVirtualSpeakers::GetVirtualSpeakersAndFolddownCoeffs(SC);

    hull_points = Join( SC.speaker_locations, V );
    H.Init( hull_points );
}
```

## 6.3 Virtual Speakers

### 6.3.1 Virtual Speakers Positioning

For listening layouts with large angular spacing between the speakers, it may be impossible to create the necessary convex hull for the proper operation of vector based 3D rendering. Successful creation of a convex hull over a set of physical speakers is achieved by carefully placing additional virtual speakers. In this case, the VBP rendering is done over a set of physical and virtual speakers and the fold-down of the virtual speakers into the physical speakers is done as a post VBP processing step. The number and position of the virtual speakers and the corresponding fold-down coefficients can greatly influence the spatial impression of the rendered sounds. `PointSourceObjectRenderer` uses two methods for placing virtual speakers: `PredefinedVirtualSpeakers` and `AutoVirtualSpeakers` in that order. If the output configuration can be supported by the `PredefinedVirtualSpeaker` algorithm, then `PredefinedVirtualSpeakers` are used, otherwise the `AutoVirtualSpeaker` algorithm is used.

### 6.3.2 Predefined Virtual Speakers

#### 6.3.2.1 Predefined Virtual Speakers Layouts

Predefined virtual speakers have been carefully designed for specific sets of physical speaker layouts. The supported sets of physical speaker layouts are classified into groups which are enumerated in clauses 7.3 and 7.4. The predefined virtual speakers consist of 10 predefined locations, a subset of which is chosen depending on the physical speaker layout.

The predefined virtual speakers in the equatorial plane are located as follows:

$$V_e = [(-135^\circ, 0^\circ, 1), (135^\circ, 0^\circ, 1)]$$

The predefined virtual speakers in the upper hemisphere are located as follows:

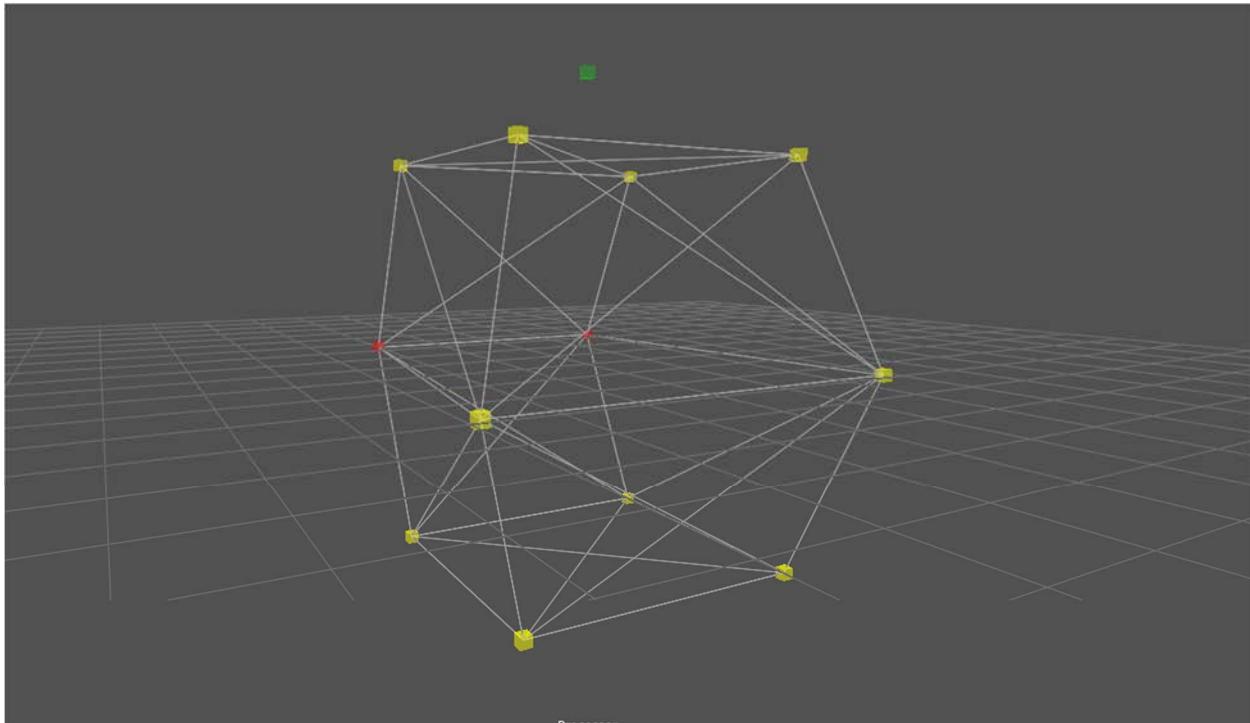
$$V_u = [(-45^\circ, 45^\circ, 1), (45^\circ, 45^\circ, 1), (-135^\circ, 45^\circ, 1), (135^\circ, 45^\circ, 1)]$$

The predefined virtual speakers in the lower hemisphere are located as follows:

$$V_l = [(-45^\circ, -45^\circ, 1), (45^\circ, -45^\circ, 1), (-135^\circ, -45^\circ, 1), (135^\circ, -45^\circ, 1)]$$

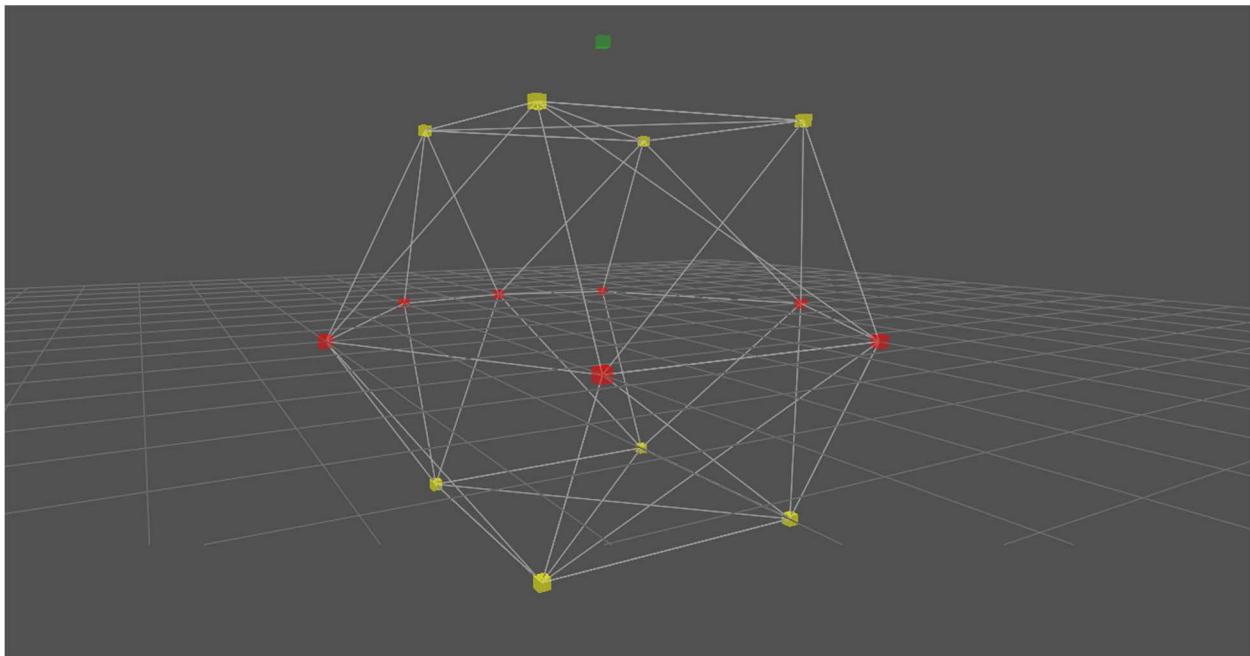
The actual virtual speakers chosen are subsets of these speakers and vary depending on the physical speaker layout. The subsets for the equatorial plane and upper hemisphere are given in clause 7.3, and for the lower hemisphere in clause 7.4.

Figure 6 shows how the virtual speakers are added for a stereo (LR) configuration. Predefined virtual speakers (yellow vertices) for the stereo destination layout (red vertices). The figure shows both the upper and lower hemisphere. The physical stereo speakers are indicated by the red vertices, and the virtual speakers in yellow are added in the equatorial plane, the upper hemisphere, and the lower hemisphere. For the stereo configuration, all ten of the possible predefined virtual speakers are used.



**Figure 6: Stereo (LR) Configuration with Predefined Virtual Speakers**

Figure 7 shows the virtual speaker setup for the C+L+R+Lss+Rss+Lsr+Rsr configuration. The figure displays virtual speakers in both the upper and lower hemispheres with yellow vertices. For this configuration, all the possible predefined virtual speakers in the upper and lower hemisphere are used.



**Figure 7: 7.x Output Configuration with Predefined Virtual Speakers**

The fold-down coefficients for predefined virtual speakers vary depending on the output layout configuration. The predefined fold-down coefficients are tabulated in clause 7.3 for the equatorial plane and upper hemisphere, and in clause 7.4 for the lower hemisphere.

### 6.3.2.2 PredefinedVirtualSpeakers::IsSupported

The pseudocode below determines whether the `PredefinedVirtualSpeakers` algorithm can support the specified speaker configuration.

```
PredefinedVirtualSpeakers::IsSupported(SpeakerConfiguration SC)
{
    // Extract channel layout mask from speaker configuration
    speaker_layout_channel_mask = SC.GetChannelMask();

    // Mask away channels not needed for equatorial plane and upper hemisphere.
    channel_mask_for_eqtr_uphemi =
        GetChannelMaskForGroups(speaker_layout_channel_mask);

    for ( group_eqtr_uphemi : list_of_all_groups_equator_or_upper )
    {
        if (group_eqtr_uphemi.list_of_channel_masks.exist(channel_mask_for_eqtr_uphemi))
        {
            return true; // Supported if the channel mask exists
        }
    }
    return false; // Unsupported
}
```

### 6.3.2.3 PredefinedVirtualSpeakers::GetVirtualSpeakersAndFolddownCoeffs

The pseudocode below shows the logic of finding the virtual speakers and their fold-down coefficients. The input to the function is the physical speaker configuration.

```
PredefinedVirtualSpeakers::GetVirtualSpeakersAndFolddownCoeffs(
    SpeakerConfiguration config )
{
    // Extract channel layout mask from speaker configuration
    speaker_layout_channel_mask = config.GetChannelMask();
    // Extract speaker locations from speaker configuration
    speaker_locations = config.speaker_locations;

    channel_mask_for_eqtr_uphemi =
        GetChannelMaskForGroups(speaker_layout_channel_mask);
```

```

// Find virtual sources and fold-down coefficients for equatorial and upper hemisphere.
for ( group_eqtr_uphemi : list_of_all_groups_equator_or_upper )
{
    if ( group_eqtr_uphemi.list_of_channel_masks.exist(channel_mask_for_eqtr_uphemi) )
    {
        V_e = group_eqtr_uphemi.eqtr_virt_srcs;
        V_u = group_eqtr_uphemi.uphemi_virt_srcs;

        FD_e = group_eqtr_uphemi.GetLinearGainMatrixEqtr( speaker_locations );
        FD_u = group_eqtr_uphemi.GetLinearGainMatrixUpHemi( speaker_locations );
    }
}

channel_mask_for_lower_hemi =
    GetChannelLayoutMaskForLowerHemisphere(physical_speaker_layout_channel_mask);

// Find virtual sources and fold-down coefficients for lower hemisphere.
for ( group_lohemi : list_of_all_groups_lower )
{
    if ( group_lohemi.list_of_channel_masks.exist(channel_mask_for_lower_hemi) )
    {
        V_l = group_lohemi.lohemi_virt_srcs;

        FD_l = group_lohemi.GetLinearGainMatrixLoHemi( speaker_locations );
    }
}

// Creating a combined list for virtual speakers
V = Join( V_e, V_u, V_l );

// Creating a combined list of fold-down coefficients
FD = Join( FD_e, FD_u, FD_l );

// return both the list of virtual speakers and their fold-down coefficients
return (V, FD);
}

```

### 6.3.3 Auto Virtual Speakers

#### 6.3.3.1 Overview of Automatic Virtual Speaker Mapping

When the SpeakerConfiguration is not supported by the PredefinedVirtualSpeaker algorithm, then the AutoVirtualSpeaker algorithm is employed. There are multiple stages in this algorithm:

- Add virtual speakers in the equatorial plane
- Add virtual speakers in the upper hemisphere
- Add virtual speakers in the lower hemisphere
- Calculate fold-down coefficients for equatorial virtual speakers
- Calculate fold-down coefficients for the upper and lower hemisphere

#### 6.3.3.2 Add Virtual Speakers in the Equatorial Plane

Virtual speakers are added in the equatorial plane if they are spaced more than 180 degrees apart when measured in the clockwise direction. They are added at 1/3 and 2/3 of the azimuth gap between the furthest speakers.

```

GetEquatorialVirtualSpeakers(SpeakerConfiguration SC)
{
    equatorial_points = GetEquatorialPoints(SC.speaker_locations);

    // Map azimuth to be between 0 and 360
    equatorial_points = MapAzimuthTo0_360(equatorial_points);

    // Sort in the ascending order of the azimuth.
    // Since elevation = 0 and radius = 1 for all points in the equator
    // of a unit sphere, the sort will order according to azimuth.
    sorted_equatorial_points = Sort(equatorial_points);
    // Repeat the first point
    sorted_equatorial_points += sorted_equatorial_points[0];
}

```

```

// Find the maximum azimuth gap and the point that has this gap.
max_gap = 0;
max_gap_p = NULL;
for ( i = 1; i < sorted_equatorial_points.size(); ++i )
{
    gap = sorted_equatorial_points[i].azimuth - sorted_equatorial_points[i-1].azimuth;
    if ( max_gap < gap )
    {
        max_gap = gap;
        max_gap_p = sorted_equatorial_points[i-1];
    }
}

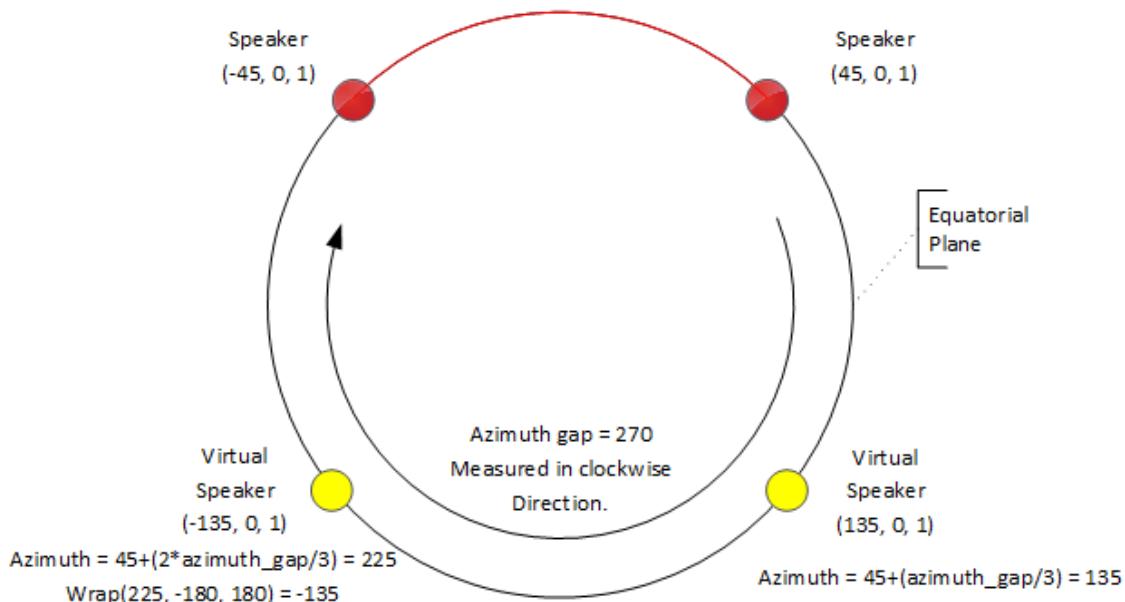
// If the gap is larger than 180, add virtual speakers in equatorial plane
// at 1/3rd and 2/3rd points between the furthest speakers.
V_e = [];

if ( max_gap > 180 )
{
    // wraps around if the angle goes beyond the range of -180 to 180.
    one_third_az = Wrap( max_gap_p + max_gap / 3.0, -180, 180 );
    two_third_az = Wrap( max_gap_p + 2 * max_gap / 3.0, -180, 180 );

    // Add virtual speaker at the 1/3rd gap
    V_e += { one_third_az, 0, 1 };
    // Add virtual speaker at the 2/3rd gap
    V_e += { two_third_az, 0, 1 };
}
return V_e;
}

```

Figure 8 illustrates the process of adding virtual speakers at 1/3rd and 2/3rd of the azimuth gap in the equatorial plane.



**Figure 8: Adding Virtual Speakers in Equatorial Plane**

### 6.3.3.3 Add Virtual Speakers in the Upper Hemisphere

Virtual speakers are added in the upper hemisphere only if there are no speakers with elevation above 0 degrees. If this condition is met, add a virtual speaker at an elevation of 45 degrees above every virtual and physical speaker in the equatorial plane.

```

GetUpperHemiVirtualSpeakers(SpeakerConfiguration SC, Point V_e[])
{
    equatorial_points = GetEquatorialPoints(SC.speaker_locations);

    // Combine both virtual and physical speakers in equatorial plane
    all_equatorial_points = Join( V_e, equatorial_points );
}

```

```

// If there are no upper hemisphere speakers, add virtuals above every
// equatorial plane speaker
V_u = []
if ( NumUpperHemispherePoints(SC.speaker_locations) == 0 )
{
    for ( p : all_equatorial_points )
    {
        V_u += { p.azimuth, 45.0, p.radius };
    }
}
return V_u;
}

```

### 6.3.3.4 Add Virtual Speakers in the Lower Hemisphere

Virtual speakers are added in the lower hemisphere only if there are no speakers with elevation above 0 degrees. If this condition is met, add a virtual speaker below every virtual and physical speaker in the equatorial plane at an elevation of -45 degrees.

```

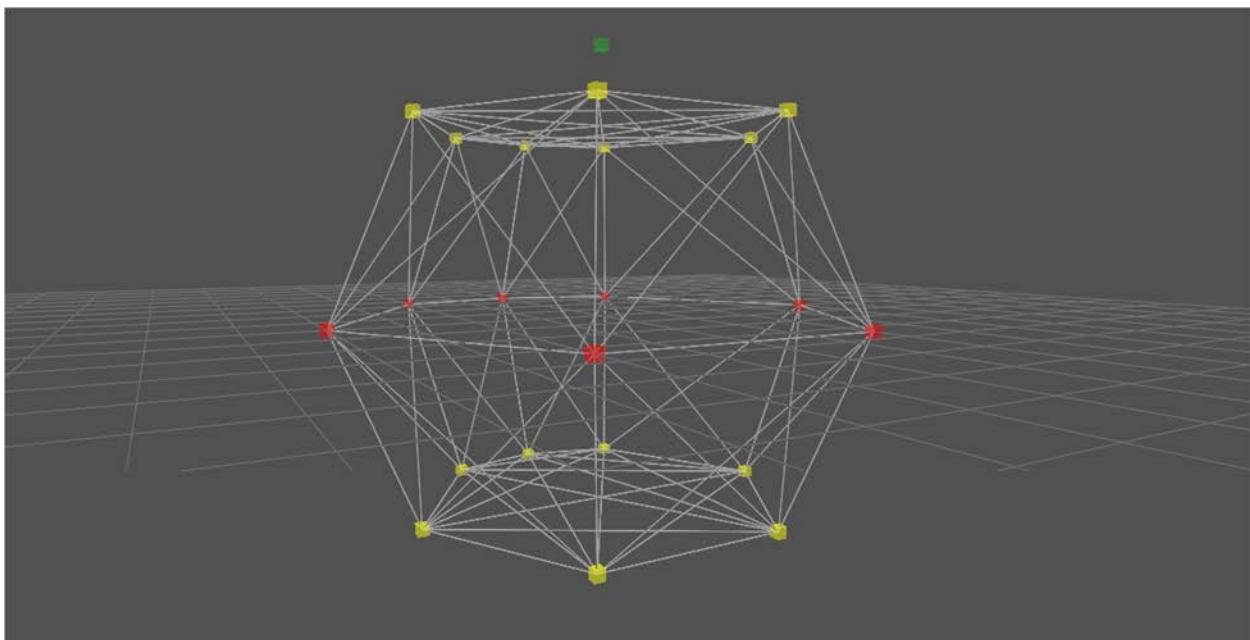
GetLowerHemiVirtualSpeakers(SpeakerConfiguration SC, Point V_e[])
{
    equatorial_points = GetEquatorialPoints(SC.speaker_locations);

    // Combine both virtual and physical speakers in equatorial plane
    all_equatorial_points = Join( V_e, equatorial_points );

    // If there are no lower hemisphere speakers, add virtuals below every
    // equatorial plane speaker
    V_l = []
    if ( NumLowerHemispherePoints(SC.speaker_locations) == 0 )
    {
        for ( p : all_equatorial_points )
        {
            V_l += { p.azimuth, 45.0, p.radius };
        }
    }
    return V_l;
}

```

Figure 9 shows the auto virtual speaker setup for the 7.x (C+L+R+Lss+Rss+Lsr+Rsr) output configuration. The yellow vertices indicate virtual speakers and the red vertices are the physical speakers.



**Figure 9: 7.x Output Configuration with Automatic Virtual Speakers**

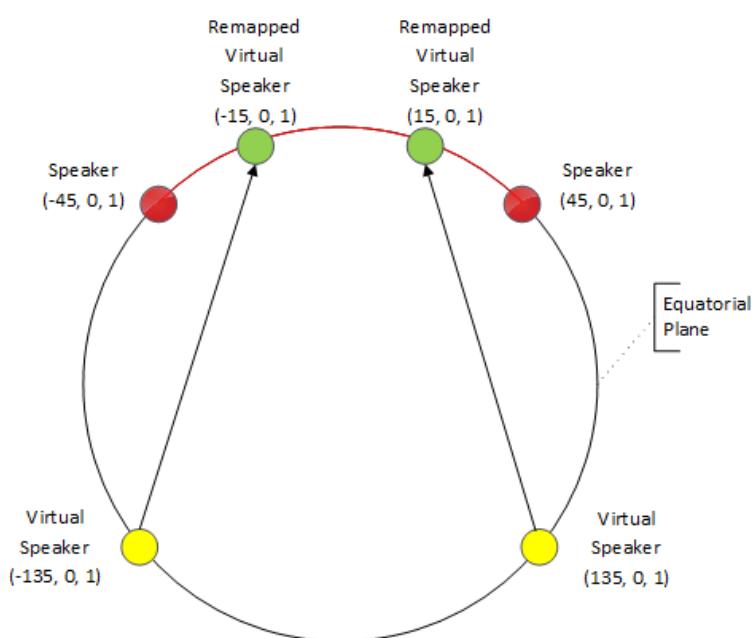
### 6.3.3.5 Calculate Fold-down Coefficients for Equatorial Virtual Speakers

If there are virtual speakers in the equatorial plane which were added at 1/3 and 2/3 of the azimuth gap between the furthest speakers, they are remapped between those physical speakers on the smaller arc on the equatorial circle and then panned pairwise to the new location for folding.

Remapping the azimuth of the virtual speaker between the physical speakers is done as shown in the pseudocode below:

```
// Remap equatorial virtual speaker (virt_az) between two furthest physical equatorial speakers
// represented by min_az and max_az.
// The azimuths angles used for this function is wrapped to be between 0 to 360 (see
MapAzimuthTo0_360 for mapping logic).
RemapEquatorialVirtualSpkr (float virt_az, float min_az, float max_az )
{
    diff_az = max_az - min_az;
    return max_az - (((virt_az - max_az) / (360.0 - diff_az)) * diff_az);
}
```

Figure 10 shows the remapping of the virtual speakers (yellow) into the arc between physical speakers (red). The remapped positions are in green.



**Figure 10: Remapping Virtual Speakers Between Physical Speakers**

The algorithm for finding the fold-down coefficients for equatorial virtual speakers is shown below. Note that a hull object is used to determine the fold-down coefficients. A temporary hull object is constructed as shown in clause 6.3.4 to calculate the fold-down coefficients.

```
GetEquatorialFoldDown(SpeakerConfiguration SC, Point V_e[], Hull H)
{
    FD_e = [][];

    // Calculate fold-down coefficients only if we have equatorial virtual speakers.
    if (V_e.size() > 0)
    {
        equatorial_points = GetEquatorialPoints(SC.speaker_locations);

        // Sort in the ascending order of the azimuth.
        // Since elevation = 0 and radius = 1 for all points in the equator
        // of a unit sphere, the sort will order according to azimuth.
        sorted_equatorial_points = Sort(equatorial_points);
        // Repeat the first point
        sorted_equatorial_points += sorted_equatorial_points[0];

        // Find the maximum azimuth gap and the points that have this gap.
        max_gap = 0;
        max_gap_p1 = NULL;
        max_gap_p2 = NULL;
        for ( i = 1; i < sorted_equatorial_points.size(); ++i )
```

```

    {
        gap = sorted_equatorial_points[i].azimuth - sorted_equatorial_points[i-1].azimuth;
        if ( max_gap < gap )
        {
            max_gap = gap;
            max_gap_p1 = sorted_equatorial_points[i-1];
            max_gap_p2 = sorted_equatorial_points[i];
        }
    }

    // Calculate the folding coefficients
    for ( p : V_e )
    {
        p_tmp = MapAzimuthTo0_360(p);
        min_plp2 = min( max_gap_p1, max_gap_p2 );
        max_plp2 = max( max_gap_p1, max_gap_p2 );
        // Find the remapped azimuth
        remap_az = RemapEquatorialVirtualSpkr ( p_tmp.azimuth, min_plp2, max_plp2 );

        // Pan the new point
        new_p = { remap_az, p.elevation, p.radius };
        gain_v = H.Pan( new_p );

        // Extract gain coefficients for the physical speakers
        FD_e += gain_v[SC.speaker_locations];
    }
}
return FD_e;
}

```

### 6.3.3.6 Calculate Fold-down Coefficients for the Upper and Lower Hemisphere

The fold-down coefficients for folding the upper and lower hemisphere virtual speakers into a speaker configuration is calculated by first collapsing the virtual speakers into the equator with a small spread. The virtual speakers are dropped down (or pulled up) by moving every point in the upper or lower hemisphere virtual speakers to a new point at elevation equal to 0 degrees. A spread is added by splitting each point into 3 points on the equator with the new azimuth = { $\theta$ ,  $\theta + 45^\circ$ ,  $\theta - 45^\circ$ }. Each of these are further scaled by {-3 dB, 0 dB, -3 dB}. This logic of calculating fold-down coefficients for the upper and lower hemisphere is illustrated in the pseudocode below. Note that a hull object is used to determine the fold-down coefficients. A temporary hull object is constructed to calculate the fold-down coefficients, as shown in clause 6.3.4.

```

GetNonEquatorialFolddown( SpeakerConfiguration SC, Point V_ul[], Point V_e float FD_e[][][], Hull H)
{
    // Same algorithm works for both upper and lower hemisphere virtual speakers.
    FD_ul = [];

    // Calculate fold-down only if virtual speakers exist
    if (V_ul.size() > 0)
    {
        float spread_angle[] = {-45, 0, 45};
        float spread_gain[] = {-3.0, 0.0, -3.0};

        for ( v : V_ul )
        {
            FD_v = [0] // 0 vector of size H.hull_points.size();
            for ( i : range(spread_angle.size()) )
            {
                p = { v.azimuth + spread_angle[i], 0, v.radius };
                // Calculate the hull gain vector
                hull_g = H.Pan(p);
                // Scale hull gain vector by the spread_gain
                mod_hull_g = hull_g * db2mag( spread_gain[i] );
                // Merge the hull gain with FD_v
                for ( h : H.hull_points )
                {
                    FD_v[h] = sqrt( pow( FD_v[h], 2 ) + pow( mod_hull_g[h], 2 ) );
                }
            }
            // The above folds virtual speakers from either
            // hemisphere to equatorial plane.
            // Now, fold the equatorial virtual speakers to physical speakers.
            for ( ve : V_e )
            {
                gain_ve = FD_v[ve]; // Extract gain contribution to 've'
                fd_ve = FD_e[ve]; // Extract fold-down vector for virtual speaker 've'
            }
        }
    }
}

```

```

        for ( sl : SC.speaker_locations )
            FD_v[sl] = FD_v[sl] + gain_ve * fd_ve[sl];
    }
    // Add the extracted gain coefficient to fold-down matrix
    FD_ul += FD_v[SC.speaker_locations];
}
}

return FD_ul;
}

```

### 6.3.4 GetVirtualSpeakersAndFolddownCoeffs

The multiple steps described in clause 6.3.3.1 are combined in the implementation of `GetVirtualSpeakersAndFolddownCoeffs()`. A temporary hull object is created using the generated virtual speakers and the speaker configuration.

```

AutoVirtualSpeakers::GetVirtualSpeakersAndFolddownCoeffs(SpeakerConfiguration SC)
{
    // Get equatorial virtual speakers
    V_e = GetEquatorialVirtualSpeakers(SC);

    // Get upper hemisphere virtual speakers
    V_u = GetUpperHemiVirtualSpeakers(SC, V_e);

    // Get lower hemisphere virtual speakers
    V_l = GetLowerHemiVirtualSpeakers(SC, V_e);

    // List of all virtual points
    V = Union( V_e, V_u, V_l );

    // Create a temporary hull for finding fold-down coefficients
    hull_points = Union( V, SC.speaker_locations );
    Hull H = Hull( hull_points );

    // Get fold-down for equatorial virtual speakers
    FD_e = GetEquatorialFolddown(SC, V_e, H);

    // Get fold-down for upper hemisphere virtual speakers
    FD_u = GetNonEquatorialFolddown(SC, V_u, V_e, FD_e, H);

    // Get fold-down for lower hemisphere virtual speakers
    FD_l = GetNonEquatorialFolddown(SC, V_l, V_e, FD_e, H);

    // Combine all fold-down coefficients
    FD = Union( FD_e, FD_u, FD_l );

    return (V,FD);
}

```

### 6.3.5 Hull Initialization

Hull points are formed by combining physical and virtual speaker locations. At the initialization stage, a set of triplets are calculated, forming the surface mesh using the hull points. An exhaustive search is performed on all the triplets formed from the hull points in order to triangulate the hull. Each triplet is associated with an area on the positively oriented surface of the sphere. The positively oriented surface is the one facing away from the origin of the sphere. This triangulation is called hull initialization and is described below in the pseudocode. To summarize the pseudocode, it is finding all the triplets that do not have another speaker point within their associated surface. It does this by checking if there is another point `p_l` inside the surface area formed by `p_i`, `p_j` and `p_k`; and if no such point exists, then the triplet { `p_i`, `p_j`, `p_k` } is on the surface of the sphere and it is saved.

```

Hull::Init(Point hull_points[])
{
    // epsilon is used to handle floating point errors.
    // A generic epsilon value is used here, but should be changed depending on platform.
    epsilon = 1e-6;

    this->hull_points = hull_points;

    // Perform an exhaustive search to find the triplets
    // that form triangles of the hull.
}

```

```

triplets_processed = []
for ( p_i : hull_points )
{
    for ( p_j : hull_points[i+1, ...] )
    {
        for ( p_k : hull_points[j+1, ...] )
        {
            // Cannot form a triangle with two same points
            if ( p_i == p_j or p_j == p_k or p_i == p_k )
                continue;

            // Construct a triplet
            Triplet t = { p_i, p_j, p_k };
            // Convert to be positively oriented
            t.ConvertToPositiveOrientation();

            // If the triplet is already processed, move on to the next triplet.
            if ( triplets_processed.contains(t) )
                continue;

            // Avoid spanning hull's facets spanning hemispheres
            if ( !t.FullyInUpperHemisphere() && !t.FullyInLowerHemisphere() )
                continue;

            Matrix M = t.GetTripletMatrix();
            // Avoid singular/degenerate matrices
            if ( !M.IsInvertible() )
                continue;

            Matrix M_inv = M.Invert();

            is_facet = true;
            for ( p_l : hull_points )
            {
                // Test vector should also be in the same hemisphere as the triplet
                if ( !t.FullyInEquatorialPlane() &&
                    t.FullyInUpperHemisphere() && p_l.elevation < 0 )
                    continue;
                if ( !t.FullyInEquatorialPlane() &&
                    t.FullyInLowerHemisphere() && p_l.elevation > 0 )
                    continue;

                g = M_inv * p_l.Cartesian();

                // This checks if the point p_l exists outside of the plane formed
                // by p_i, p_j & p_k away from origin. The sum of gains will be > 1
                // only for a point lying above the plane but within the triangle
                // formed by p_i, p_j & p_k.
                if ( g[0] + g[1] + g[2] > 1 + epsilon )
                {
                    is_facet = false;
                    break;
                }
            }

            // If the triplet is a face, add it hull triplets
            if ( is_facet )
                this->T += t;

            // Mark triplet as processed
            triplets_processed += t;
        } // for p_k
    } // for p_j
} // for p_i
}

```

## 6.4 Predefined Virtual Speakers Organization

The predefined virtual speaker assignments and corresponding fold-down coefficients are organized by the channel bitmask. There are 21 non-overlapping groups of channel bitmasks for the equatorial plane and/or upper hemisphere regions that have defined virtual speakers with corresponding fold-down coefficients. These groups are defined in clause 7.3.

A subset of the channel masks that include speakers in the equatorial plane also have predefined virtual speakers in the lower hemisphere. This subset is further organized into five non-overlapping groups. These groups are defined in clause 7.4.

## 6.5 Rendering All Objects

### 6.5.1 Object Rendering

Rendering all objects involves panning all the objects in the specified SpeakerConfiguration, followed by smoothing and mixing them into the output feeds.

The pseudocode below illustrates the implementation of `RenderAllObjects()`, which renders one `AudioObject` at a time. Similarly, `RenderAudioObject()` renders one `MonoObject` at a time.

```
PointSourceObjectRenderer::RenderAllObjects(AudioObject AO_list[], uint block_size)
{
    // Prepare output buffers for rendering
    out_waveforms = float[SC.speaker_locations.size()][block_size];
    for (AudioObject AO : AO_list)
    {
        RenderAudioObject(AO, block_size);
    }
    return out_waveforms;
}

PointSourceObjectRenderer::RenderAudioObject(AudioObject AO, uint block_size)
{
    for (MonoObject MO : AO.mono_objects)
    {
        RenderMonoObject(MO, block_size, AO.obj_gain, AO.obj_lambda);
    }
};
```

A mono object is defined as a single waveform with an associated list of point source locations. To render a mono object, each of the point source locations is panned into the speaker configuration. The gain vectors of all the points are summed together to get a waveform's gain vector. It is then smoothed with the previous block's gain vector and applied on the waveform to render it into the output feeds. The pseudocode for `RenderMonoObject()` is below.

```
PointSourceObjectRenderer::RenderMonoObject(
    MonoObject MO,
    uint block_size,
    float obj_gain,
    float obj_lambda)
{
    // Initialize MonoObject gain array.
    float MO_gains[SC.speaker_locations.size()];

    for ( Point p : MO.positions )
    {
        // Calculate gains from the panner.
        float gains[] = VBP.Pan(p);
        for ( Point sl : SC.speaker_locations )
        {
            // Apply object gain to the VBP gains and accumulate into the MonoObject's gain.
            MO_gains[sl] += gains[sl] * obj_gain;
        }
    }
    for ( Point sl : SC.speaker_locations )
    {
        // set dynamic lambda for the smoothers.
        MO.smoothening_list[sl].lambda = obj_lambda;
        // Run the smoothing filter.
        MO_smoothed_gain_over_a_block = MO.smoothening_list[sl].Smooth(MO_gains[sl], block_size);
        for ( uint t : 0 to block_size )
        {
            // Apply smoothed gains to the MonoObject waveforms to render.
            out_waveforms[sl][t] += MO_smoothed_gain_over_a_block[t] * MO.waveform[t] * obj_gain;
        }
    }
}
```

## 6.5.2 Vector Based Panner

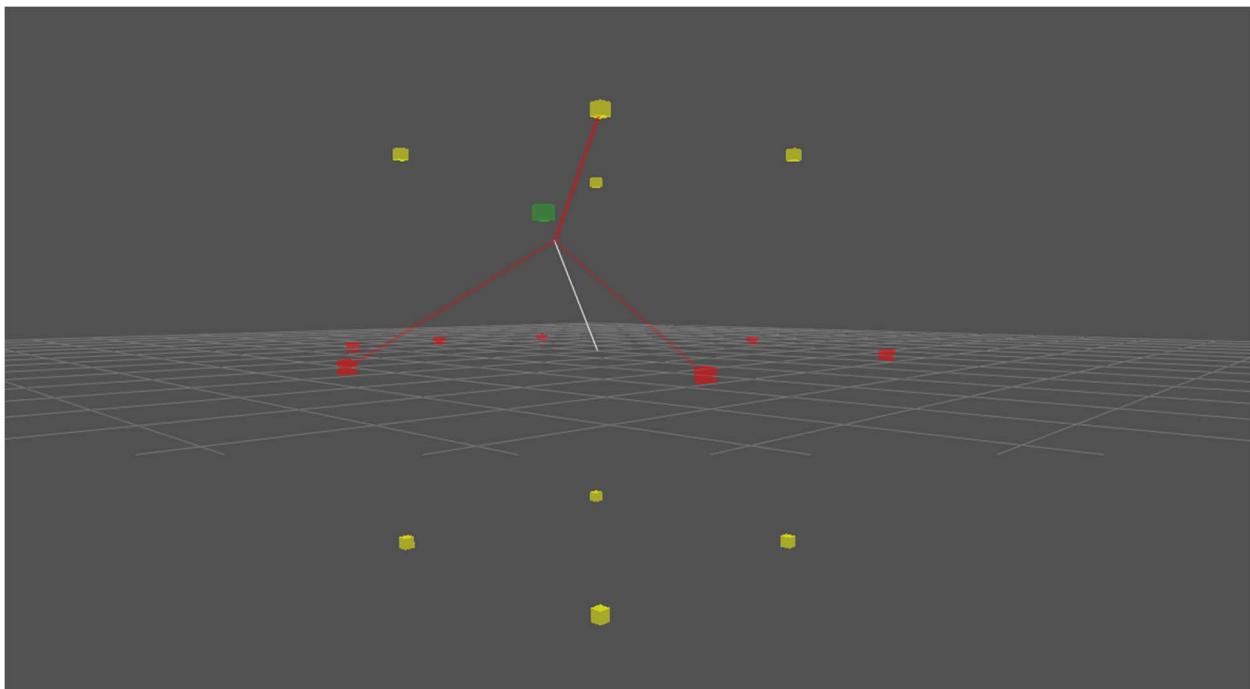
The vector based panner takes a point and pans it to the speaker configuration. VBP manages the hull, virtual speakers and their fold-down coefficients. The panning algorithm first pans the point to the hull and then fold-down coefficients are applied to the panned gain coefficients in the specified speaker configuration. The gain coefficients are normalized to preserve power. The pseudocode below implements this algorithm.

```
VectorBasedPanner::Pan(Point P)
{
    // Pan to hull
    gain_H = H.Pan(p);

    // Fold-down
    gain = [0...SC.speaker_locations] = 0; // zero vector
    for ( v : V ) // for each virtual speaker
    {
        gv = gain_H[v];
        for ( sl : SC.speaker_locations )
        {
            gain[sl] += gv * FD[v][sl];
        }
    }
    // normalize
    gain /= sqrt( gain.Dot(gain) );
    return gain;
}
```

## 6.5.3 Panning in a Hull

After the hull triplets are constructed (see clause 6.3.5), the panner is ready to pan any point on the surface of the sphere. For a given Hull, the panner renders a point source location in that hull by first finding set of triplets that contain the point source location and then pan the point in those triplets using VectorBasedPanner. The renderer uses vector based amplitude panning, which extends the well-known tangent law for pairwise panning on a plane to three-dimensions (see [2]). Figure 11 shows panning of a point source location into 7.x (C+L+R+Lss+Rss+Lsr+Rsr) speaker configuration. The red lines are the contribution line to each of the point in the triplet. The point source location (green vertex) is panned to a triplet within the 7.x output configuration. The white line is a vector in the direction of the point source from the origin.

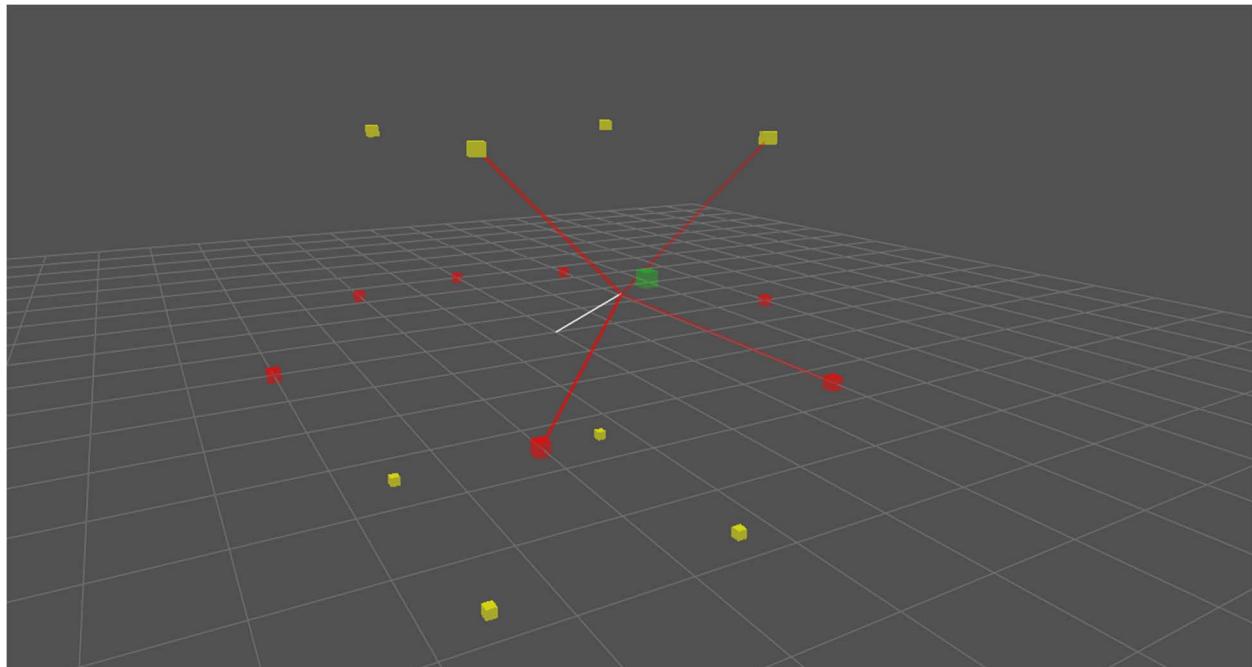


**Figure 11: 7.x Speaker Configuration Point Source Location Panning**

Multiple triplets can contain the same point source locations because of the following conditions:

- The point may lie on a speaker where it belongs to all the triplets formed with that speaker; or
- The point may be on the edge between two triplets; or
- The point may be inside an n-patch, a three-dimensional polygon that has overlapping triangles.

To address such cases, all the triplets are used and the gains are averaged over all the triplets. Figure 12 shows how the panner distributes gains for a point source location which spans over multiple triplets. The point source location (green vertex) is panned to a 4-patch within the 7.x output configuration.



**Figure 12: Point Source Panner Distribution of Gains**

When a point is located on the edge between two triplets, in addition to the two triplets sharing the edge, it is possible that there may be a third triplet overlapping these two triplets fully containing the point. In such cases, it is necessary to prevent the speakers on the edge from getting a boost over the other speakers. That is, instead of computing the gain as  $gain = (2 * edge + triplet)/3$  it is calculated as  $gain = (edge + triplet)/2$ . In this manner, a moving point source location has continuous gain changes. To handle this, the panning algorithm treats edge cases differently by storing them and ensuring that other triplets in the edge will not be used for panning.

Note that the panning pseudocode below uses a generic value for `epsilon`, initialized in `Hull::Init()`, to handle the floating point errors. An actual implementation may need to change this value or handle floating point errors differently.

Given a point, panning to the hull is implemented in the pseudocode below.

```
Hull::Pan(Point p)
{
    // gain vector for all points in the hull; initialized to 0
    gain_H = [0...hull_points.size()];

    // List of all the significant edges that were panned.
    all_significant_edge_list = [];

    // List of all the edges already panned.
    all_panned_edge_list = [];

    triplet_count = 0;
    for (t : T) // For all triplets
    {
        Matrix M_inv = t.GetTripletMatrix().Inverse();

        // Calculate gain vector by panning p to this triplet
        Vector g = M_inv * p.Cartesian();
```

```

// Check if the point p lies in this triplet
if ( g[0] >= -epsilon && g[1] >= -epsilon && g[2] >= -epsilon )
{
    if (g[0] < 0) g[0] = 0;
    if (g[1] < 0) g[1] = 0;
    if (g[2] < 0) g[2] = 0;

    significance_count = ( g[0] > epsilon ) + ( g[1] > epsilon ) + ( g[2] > epsilon );

    // Special case when the point lies on the edge of a triangle.
    // Since this edge would appear in two triangles, doubling the contribution
    // of the edge. To avoid this, we ensure that a significant edge is not panned twice.
    if ( significance_count == 2 ) {
        Edge significant_edge;
        if (g[0] > epsilon && g[1] > epsilon)
            significant_edge = Edge(t.p1, t.p2);
        else if (g[0] > epsilon && g[2] > epsilon)
            significant_edge = Edge(t.p1, t.p3);
        else
            significant_edge = Edge(t.p2, t.p3);

        // If the significant edge is already panned then move on to the next triplet.
        if ( all_panned_edge_list.exist(significant_edge) )
            continue;

        // Add significant edge to the special list
        all_significant_edge_list += significant_edge;
    }

    // Triplets' edges
    Edge triplet_edge_list[] = { Edge(t.p1, t.p2), Edge(t.p1, t.p3), Edge(t.p2, t.p3) };

    // Special case when any of the edge has been already marked as significant.
    // This would occur due floating point errors and hence, once an edge is marked
    // significant, it remains significant. And so, we shall render that edge only once.
    edge_is_significant_and_panned = false;
    for ( Edge e : triplet_edge_list )
        if (all_significant_edge_list.exist(e) && all_panned_edge_list.exist(e)) {
            edge_is_significant_and_panned = true;
            break;
        }
    if ( edge_is_significant_and_panned )
        continue; // if edge is significant and already panned, move on to next triplet.

    // Add the edges in the triplet to the list of panned edges.
    all_panned_edge_list += triplet_edge_list;

    // Add the triplet contribution to hull gain
    gain_H[ t.p1 ] += g[0];
    gain_H[ t.p2 ] += g[1];
    gain_H[ t.p3 ] += g[2];
    ++triplet_count;
}
}

// Normalize the gains
gain_H /= triplet_count;

return gain_H;
}

```

## 7 Predefined Virtual Speaker Mapping

### 7.1 Channel Bitmask

The channel layouts defined in clause 7.3 and clause 7.4 contain the entire set supported by the predefined virtual speaker mode. The channel masks are specified by the channel layout bitmask which is formed by taking bitwise disjunction of individual channels' bitmasks as defined in Table 3.

Note that there is some exclusion of bitmasks while listing the channel masks. For example, in the clauses of 7.3, the LFE channels are not included in the channel mask listing. In clause 7.4, the LFE channels and all of the upper hemisphere channels are excluded from the channel mask listing. See clause 7.3.1.1 and clause 7.4.2.1 for the channel mask conversion functions.

**Table 3: Channel Definitions and Bitmask**

Channel Description	Abbreviation	Azimuth (degrees)	Elevation (degrees)	Channel Bitmask
Centre	C	0,0	0,0	0x00000001
Left	L	-30,0	0,0	0x00000002
Right	R	30,0	0,0	0x00000004
Left Surround	Ls	-110,0	0,0	0x00000008
Right Surround	Rs	110,0	0,0	0x00000010
Low Frequency Effects - 1	LFE1	0,0	0,0	0x00000020
Centre Surround	Cs	180,0	0,0	0x00000040
Left Surround Rear	Lsr	-150,0	0,0	0x00000080
Right Surround Rear	Rsr	150,0	0,0	0x00000100
Left Side Surround	Lss	-90,0	0,0	0x00000200
Right Side Surround	Rss	90,0	0,0	0x00000400
Left Centre	Lc	-15,0	0,0	0x00000800
Right Centre	Rc	15,0	0,0	0x00001000
Left High	Lh	-30,0	45,0	0x00002000
Centre High	Ch	0,0	45,0	0x00004000
Right High	Rh	30,0	45,0	0x00008000
Low Frequency Effects - 2	LFE2	0,0	0,0	0x00010000
Left Wide	Lw	-60,0	0,0	0x00020000
Right Wide	Rw	60,0	0,0	0x00040000
Overhead	Oh	0,0	90,0	0x00080000
Left High Side	Lhs	-90,0	45,0	0x00100000
Right High Side	Rhs	90,0	45,0	0x00200000
Centre high Rear	Chr	180,0	45,0	0x00400000
Left High Rear	Lhr	-150,0	45,0	0x00800000
Right High Rear	Rhr	150,0	45,0	0x01000000
Centre Low Front	Clf	0,0	-30,0	0x02000000
Left Low Front	Llf	-30,0	-30,0	0x04000000
Right Low Front	Rlf	30,0	-30,0	0x08000000
Left Top Front	Ltf	-45,0	60,0	0x10000000
Right Top Front	Rtf	45,0	60,0	0x20000000
Left Top Rear	Ltr	-135,0	60,0	0x40000000
Right Top Rear	Rtr	135,0	60,0	0x80000000

## 7.2 General Mapping Functions

### 7.2.1 GetLinearGainVectorFromTable

This function converts the fold-down coefficients tables defined in clause 7.3 and clause 7.4 into gain vectors.

```
// Get fold-down coefficients for a virtual point in linear scale.
GetLinearGainVectorFromTable(
    GainCoefficient_dB table,
    FunctionP mapping_functions[],
    uint virtual_p_idx,
    Point speaker_locations[] )
{
    out_gain_vector = []
    for ( i : range(speaker_locations.size()) ) 
        out_gain_vector[i] = 0;

    for ( func_p : mapping_functions )
        out_gain_vector[func_p()] = db2mag(table[in_virtual_p_idx][func_p()]);
}

return out_gain_vector;
}
```

## 7.2.2 Physical Speaker Search Functions

### 7.2.2.1 Search Functions Overview

A physical speaker is identified by the location of the speaker in the listening space. This location can be specified as a tuple of azimuth  $\theta$ , elevation  $\varphi$  and radius  $r$ . As with the object positions, azimuth varies from -180 degrees to 180 degrees; a negative angle indicates speaker on the left hemisphere and the positive angle indicates speaker on the right hemisphere. The elevation varies from -90 degrees to 90 degrees; where a negative angle indicates lower hemisphere and the positive angle indicates upper hemisphere. A virtual speaker is defined in the same manner as a physical speaker, but they do not physically exist in the listening space. The following functions are used to identify the physical speaker location. All of the functions listed below are using the set of physical speaker locations, `SC.speaker_locations`.

### 7.2.2.2 ClosestAzEq

`ClosestAzEq( $\theta$ )` returns the physical speaker index from the set `SC.speaker_locations[ ]` that meets two conditions:

- 1) It is located in equatorial plane ( $\varphi = 0$ ).
- 2) It is closest to the specified azimuth ( $\theta$ ).

### 7.2.2.3 ClosestAzUH

`ClosestAzUH( $\theta$ )` returns the physical speaker index from the set `SC.speaker_locations[ ]` that meets two conditions:

- 1) It is located in the equatorial plane or upper hemisphere ( $\varphi \geq 0$ ).
- 2) It is closest to the specified azimuth ( $\theta$ ).

### 7.2.2.4 ClosestElv

`ClosestElv( $\varphi$ , S)` returns the physical speaker index from the set `S`, which is a subset of `SC.speaker_locations[ ]` whose elevation is closest to the specified elevation  $\varphi$ .

### 7.2.2.5 LeftLargestEqtr

`LeftLargestEqtr()` returns the physical speaker index from set `SC.speaker_locations[ ]` that meets two conditions:

- 1) It is located on the equatorial plane ( $\varphi = 0$ ).
- 2) It has largest magnitude azimuth in the left hemisphere ( $0 > \theta > -180^\circ$ ).

### 7.2.2.6 RightLargestEqtr

`RightLargestEqtr()` returns the physical speaker index from set in `SC.speaker_locations[ ]` that meets two conditions:

- 1) It is located on the equatorial plane ( $\varphi = 0$ ).
- 2) It has largest azimuth in the right hemisphere ( $0 < \theta < 180^\circ$ ).

### 7.2.2.7 PrefOrder

`PrefOrder(Spkr1, Spkr2, ...)` is used to specify the preference order of speakers to contribute to. It returns the index of the first speakers from the input ordered speaker list that exist in the physical speakers set. It is used in cases where all of `Spkr1`, `Spkr1+ Spkr2 & Spkr2` are possible output layouts. To resolve such cases, the contribution preference order is used, `Spkr1` gets the contribution if it is present or else `Spkr2` gets the contribution, and so on. Note that the function takes an arbitrary number of arguments; as such, any number of speakers may be used while calling this function. Speakers can also be identified by calling other search functions.

### 7.2.2.8 Split

`Split(Spkr1, Spkr2, ..., Gain_dB)` is used to specify the split of contribution between speakers `Spkr1, Spkr2 ... SpkrN` by `Gain_dB` gain. The splitting only happens if all the speakers in the list exist.

### 7.2.2.9 ExclusiveOr

`ExclusiveOr(Spkr1, Spkr2, ...)` is used to specify disjunction choice of the speakers. In such cases only one of the speakers exists and hence, apply the contribution specified to that speaker.

## 7.3 Equatorial Plane and Upper Hemisphere

### 7.3.1 Equatorial Plane and Height Functions

#### 7.3.1.1 GetChannelMaskForEqtrUpHemiGroups

This function masks the LFE channels to zero from the physical channel layouts speakers.

```
GetChannelMaskForEqtrUpHemiGroups( physical_speaker_channel_layout_mask )
{
    // Defining masking flags that will be masked to zero for channel layout groups.
    mask_lfe = LFE1 | LFE2;

    // Take negation of lfe masks
    mask_to_zero = (~mask_lfe);

    // Apply conjunction to make lfe's channel masks zero.
    channel_layout_mask_for_groups =
        physical_speaker_channel_layout_mask & mask_to_zero;

    // return the resultant channel mask.
    return channel_layout_mask_for_groups;
}
```

#### 7.3.1.2 GroupEquatorOrUpper

The following structure represents a group in the equatorial plane or the upper hemisphere virtual speakers.

```
struct GroupEquatorOrUpper
{
    // An array listing all the channel masks supported by the group.
    uint32 list_of_channel_masks[];

    // An array of equatorial virtual speakers used by the group.
    Point eqtr_virt_srcs[];
    // An array of upper hemisphere virtual speakers used by the group.
    Point uphemi_virt_srcs[];

    // An array of function pointers to determine physical speaker index.
    FunctionP eqtr_functions[];
    // An array of function pointers to determine upper hemisphere virtual speaker index.
    FunctionP uphemi_functions[];

    // Table of fold-down coefficients for the equatorial virtual speakers.
    GainCoefficient_dB eqtr_folddown[][];
    // Table of fold-down coefficients for the upper hemisphere virtual speakers.
    GainCoefficient_dB uphemi_folddown[][];
```

```

// Get linear gain vector for virtual sources in equatorial plane
GetLinearGainMatrixEqtr(Point in_output_locations[]);
// Get linear gain vector for virtual sources in upper hemisphere
GetLinearGainMatrixUpHemi(Point speaker_locations[]);
}

GroupEquatorOrUpper::GetLinearGainMatrixEqtr(Point speaker_locations[])
{
    out_gain_matrix = [][];

    for ( virtual_p_idx : range(eqtr_virt_srcs.size()) )
    {
        out_gain_matrix[virtual_p_idx] =
            GetLinearGainVectorFromTable(
                eqtr_virt_srcs,
                eqtr_functions,
                virtual_p_idx,
                speaker_locations );
    }
    return out_gain_matrix;
}

GroupEquatorOrUpper:: GetLinearGainMatrixUpHemi(Point speaker_locations[])
{
    out_gain_matrix = [][];

    for ( virtual_p_idx : range(uphemi_virt_srcs.size()) )
    {
        out_gain_matrix[virtual_p_idx] =
            GetLinearGainVectorFromTable(
                uphemi_virt_srcs,
                uphemi_functions,
                virtual_p_idx,
                speaker_locations );
    }
    return out_gain_matrix;
}

```

### 7.3.1.3 list\_of\_all\_groups\_equator\_or\_upper

The following list defines all the supported groups for custom virtual speakers in the upper hemisphere and the equatorial plane.

```

list_of_all_groups_equator_or_upper = {
    GroupEquatorOrUpper1,
    GroupEquatorOrUpper2,
    ...
    GroupEquatorOrUpper21
};

```

## 7.3.2 GroupEquatorOrUpper1

### 7.3.2.1 Channel Layouts in GroupEquatorOrUpper1

Channel layout bitmasks that use GroupEquatorOrUpper1 fold-down coefficients are listed in Table 4.

**Table 4: GroupEquatorOrUpper1 list\_of\_channel\_masks[]**

0x00000006	0x00001800	0x00001E00	0x00060000	0x00060600	0x00061800	0x00061E00
0x00000007	0x00001801	0x00001E01	0x00060001	0x00060601	0x00061801	0x00061E01
0x00000606	0x00001806	0x00001E06	0x00060006	0x00060606	0x00061806	0x00061E06
0x00000607	0x00001807	0x00001E07	0x00060007	0x00060607	0x00061807	0x00061E07



### 7.3.3.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper2

Fold-down coefficients for virtual speakers in the equatorial plane are derived from Table 8.

**Table 8: GroupEquatorOrUpper2 Fold-down Coefficients for Virtual Speakers in Equatorial Plane**

Physical Speakers eqtr_functions	LeftLargestEqtr()	RightLargestEqtr()	PrefOrder(Lhr,Ltr)	PrefOrder(Rhr,Rtr)
Virtual Speakers eqtr_virt_srcs	eqtr_folddown[][]			
(-135,0,1)	-3,0103 dB		-3,0103 dB	
(135,0,1)		-3,0103 dB		-3,0103 dB

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 9.

**Table 9: GroupEquatorOrUpper2 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	ClosestAzEq(-40)	ClosestAzEq(40)	PrefOrder(Ltr,Lhr)	PrefOrder(Rtr,Rhr)
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]			
(-45,45,1)	-3,0103 dB		-3,0103 dB	
(45,45,1)		-3,0103 dB		-3,0103 dB

### 7.3.4 GroupEquatorOrUpper3

#### 7.3.4.1 Channel Layouts in GroupEquatorOrUpper3

Channel layout bitmasks that use GroupEquatorOrUpper3 fold-down coefficients are listed in Table 10.

**Table 10: GroupEquatorOrUpper3 list\_of\_channel\_masks[]**

0x00404006	0x00405800	0x00405E00	0x00464000	0x00464600	0x00465800	0x00465E00
0x00404007	0x00405801	0x00405E01	0x00464001	0x00464601	0x00465801	0x00465E01
0x00404606	0x00405806	0x00405E06	0x00464006	0x00464606	0x00465806	0x00465E06
0x00404607	0x00405807	0x00405E07	0x00464007	0x00464607	0x00465807	0x00465E07

#### 7.3.4.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper3

Fold-down coefficients for virtual speakers in the equatorial plane are derived from Table 11.

**Table 11: GroupEquatorOrUpper3 Fold-down Coefficients for Virtual Speakers in Equatorial Plane**

Physical Speakers eqtr_functions	LeftLargestEqtr()	RightLargestEqtr()	Chr
Virtual Speakers eqtr_virt_srcs	eqtr_folddown[][]		
(-135,0,1)	-3,0103 dB		-3,0103 dB
(135,0,1)		-3,0103 dB	-3,0103 dB

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 12.

**Table 12: GroupEquatorOrUpper3 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	ClosestAzEq(-40)	ClosestAzEq(40)	Ch
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]		
(-45,45,1)	-3,0103 dB		-3,0103 dB
(45,45,1)		-3,0103 dB	-3,0103 dB

### 7.3.5 GroupEquatorOrUpper4

#### 7.3.5.1 Channel Layouts in GroupEquatorOrUpper4

Channel layout bitmasks that use GroupEquatorOrUpper4 fold-down coefficients are listed in Table 13.

**Table 13: GroupEquatorOrUpper4 list\_of\_channel\_masks[]**

0x00300006	0x00301800	0x00301E00	0x00360000	0x00360600	0x00361800	0x00361E00
0x00300007	0x00301801	0x00301E01	0x00360001	0x00360601	0x00361801	0x00361E01
0x00300606	0x00301806	0x00301E06	0x00360006	0x00360606	0x00361806	0x00361E06
0x00300607	0x00301807	0x00301E07	0x00360007	0x00360607	0x00361807	0x00361E07

#### 7.3.5.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper4

Fold-down coefficients for virtual speakers in the equatorial plane are derived from Table 14.

**Table 14: GroupEquatorOrUpper4 Fold-down Coefficients for Virtual Speakers in Equatorial Plane**

Physical Speakers eqtr_functions	PrefOrder (Lss,Lhs,Ltm)	PrefOrder (Rss,Rhs,Rtm)
Virtual Speakers eqtr_virt_srcs	eqtr_folddown[][]	
(-135,0,1)	0 dB	
(135,0,1)		0 dB

**Table 15: GroupEquatorOrUpper4 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	ClosestAzEq(-40)	ClosestAzEq(40)	ExclusiveOr (Lhs,Ltm)	ExclusiveOr (Rhs,Rtm)
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]			
(-45,45,1)	-3,0103 dB		-3,0103 dB	
(45,45,1)		-3,0103 dB		-3,0103 dB

### 7.3.6 GroupEquatorOrUpper5

#### 7.3.6.1 Channel Layouts in GroupEquatorOrUpper5

Channel layout bitmasks that use GroupEquatorOrUpper5 fold-down coefficients are listed in Table 16.

**Table 16: GroupEquatorOrUpper5 list\_of\_channel\_masks[]**

0x00400006	0x00401800	0x00401E00	0x00460000	0x00460600	0x00461800	0x00461E00
0x00400007	0x00401801	0x00401E01	0x00460001	0x00460601	0x00461801	0x00461E01
0x00400606	0x00401806	0x00401E06	0x00460006	0x00460606	0x00461806	0x00461E06
0x00400607	0x00401807	0x00401E07	0x00460007	0x00460607	0x00461807	0x00461E07

#### 7.3.6.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper5

Fold-down coefficients for virtual speakers in the equatorial plane are derived from Table 17.

**Table 17: GroupEquatorOrUpper5 Fold-down Coefficients for Virtual Speakers in Equatorial Plane**

Physical Speakers eqtr_functions	LeftLargestEqtr()	RightLargestEqtr()	Chr
Virtual Speakers eqtr_virt_srcs	eqtr_folddown[][]		
(-135,0,1)	-12,3045 dB		-0,2633 dB
(135,0,1)		-12,3045 dB	-0,2633 dB

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 18.

**Table 18: GroupEquatorOrUpper5 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	ClosestAzEq(-40)	ClosestAzEq(40)
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]	
(-45,45,1)	0 dB	
(45,45,1)		0 dB

### 7.3.7 GroupEquatorOrUpper6

#### 7.3.7.1 Channel Layouts in GroupEquatorOrUpper6

Channel layout bitmasks that use GroupEquatorOrUpper6 fold-down coefficients are listed in Table 19.

**Table 19: GroupEquatorOrUpper6 list\_of\_channel\_masks[]**

0x00004006	0x00005800	0x00005E00	0x00064000	0x00064600	0x00065800	0x00065E00
0x00004007	0x00005801	0x00005E01	0x00064001	0x00064601	0x00065801	0x00065E01
0x00004606	0x00005806	0x00005E06	0x00064006	0x00064606	0x00065806	0x00065E06
0x00004607	0x00005807	0x00005E07	0x00064007	0x00064607	0x00065807	0x00065E07





















**Table 33: GroupEquatorOrUpper11 list\_of\_channel\_masks[]**

0x00000001E	0x0000181E	0x0006001E	0x0006181E
0x0000001F	0x0000181F	0x0006001F	0x0006181F

### 7.3.12.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper11

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 34.

**Table 34: GroupEquatorOrUpper11 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	L	R	Ls	Rs
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]			
(-45, 45, 1)	0,0 dB			
(45, 45, 1)		0,0 dB		
(-135, 45, 1)			-0,4576 dB	-10 dB
(135, 45, 1)			-10 dB	-0,4576 dB

Fold-down coefficients for virtual speakers in lower hemisphere are defined in clause 7.4.5.

### 7.3.13 GroupEquatorOrUpper12

#### 7.3.13.1 Channel Layouts in GroupEquatorOrUpper12

Channel layout bitmasks that use GroupEquatorOrUpper12 fold-down coefficients are listed in Table 35.

**Table 35: GroupEquatorOrUpper12 list\_of\_channel\_masks[]**

0x00000005E	0x0000185E	0x0006005E	0x0006185E
0x00000005F	0x0000185F	0x0006005F	0x0006185F

#### 7.3.13.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper12

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 36.

**Table 36: GroupEquatorOrUpper12 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	L	R	Ls	Rs	Cs
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]				
(-45, 45, 1)	0,0 dB				
(45, 45, 1)		0,0 dB			
(-135, 45, 1)			-3,0103 dB		-3,0103 dB
(135, 45, 1)				-3,0103 dB	-3,0103 dB

Fold-down coefficients for virtual speakers in lower hemisphere are defined in clause 7.4.6.

### 7.3.14 GroupEquatorOrUpper13

#### 7.3.14.1 Channel Layouts in GroupEquatorOrUpper13

Channel layout bitmasks that use GroupEquatorOrUpper13 fold-down coefficients are listed in Table 37.

**Table 37: GroupEquatorOrUpper13 list\_of\_channel\_masks[]**

0x00000019E	0x0000199E	0x0006019E	0x0006199E
0x00000019F	0x0000199F	0x0006019F	0x0006199F
0x0000001DE	0x000019DE	0x000601DE	0x000619DE
0x0000001DF	0x000019DF	0x000601DF	0x000619DF
0x000000786	0x00001F86	0x00060786	0x00061F86
0x000000787	0x00001F87	0x00060787	0x00061F87
0x0000007C6	0x00001FC6	0x000607C6	0x00061FC6
0x0000007C7	0x00001FC7	0x000607C7	0x00061FC7

#### 7.3.14.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper13

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 38.

**Table 38: GroupEquatorOrUpper13 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	L	R	Ls	Rs
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]			
(-45, 45, 1)	0,0 dB			
(45, 45, 1)		0,0 dB		
(-135, 45, 1)			0,0 dB	
(135, 45, 1)				0,0 dB

Fold-down coefficients for virtual speakers in lower hemisphere are defined in clause 7.4.7.

### 7.3.15 GroupEquatorOrUpper14

#### 7.3.15.1 Channel Layouts in GroupEquatorOrUpper14

Channel layout bitmasks that use GroupEquatorOrUpper14 fold-down coefficients are listed in Table 39.

**Table 39: GroupEquatorOrUpper14 list\_of\_channel\_masks[]**

0x0000A01E	0x0000A01F	0x3000001E	0x3000001F
------------	------------	------------	------------

#### 7.3.15.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper14

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 40.

**Table 40: GroupEquatorOrUpper14 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	Ls	Rs
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]	
(-135, 45,1)	-0,4576 dB	-10 dB
(135, 45,1)	-10 dB	-0,4576 dB

### 7.3.16 GroupEquatorOrUpper15

#### 7.3.16.1 Channel Layouts in GroupEquatorOrUpper15

Channel layout bitmasks that use fold-down coefficients in GroupEquatorOrUpper15 are defined in Table 41.

**Table 41: GroupEquatorOrUpper15 list\_of\_channel\_masks[]**

0x0030001E	0x0030001F
------------	------------

#### 7.3.16.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper15

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 42.

**Table 42: GroupEquatorOrUpper15 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	L	R	Ls	Rs	ExclusiveOr (Lhs,Ltm)	ExclusiveOr (Rhs,Rtm)
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]					
(-45, 45,1)	-3,0103 dB				-3,0103 dB	
(45, 45,1)		-3,0103 dB				-3,0103 dB
(-135, 45,1)			-4,6976 dB	-13,7828 dB	-2,0823 dB	
(135, 45,1)			-13,7828 dB	-4,6976 dB		-2,0823 dB

### 7.3.17 GroupEquatorOrUpper16

#### 7.3.17.1 Channel Layouts in GroupEquatorOrUpper16

Channel layout bitmasks that use fold-down coefficients in GroupEquatorOrUpper16 are listed in Table 43.

**Table 43: GroupEquatorOrUpper16 list\_of\_channel\_masks[]**

0x0040401E	0x0040401F
------------	------------

#### 7.3.17.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper16

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 44.

**Table 44: GroupEquatorOrUpper16 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	L	R	Ls	Rs	Ch	Chr
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[]])					
(-45, 45, 1)	-3,0103 dB				-3,0103 dB	
(45, 45, 1)		-3,0103 dB			-3,0103 dB	
(-135, 45, 1)			-3,0103 dB			-3,0103 dB
(135, 45, 1)				-3,0103 dB		-3,0103 dB

### 7.3.18 GroupEquatorOrUpper17

#### 7.3.18.1 Channel Layouts in GroupEquatorOrUpper17

Channel layout bitmasks that use GroupEquatorOrUpper17 fold-down coefficients are listed in Table 45.

**Table 45: GroupEquatorOrUpper17 list\_of\_channel\_masks[]**

0x0040A01E	0x0040A01F	0x3040001E	0x3040001F
------------	------------	------------	------------

#### 7.3.18.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper17

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 46.

**Table 46: GroupEquatorOrUpper17 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	Ls	Rs	Chr
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[]])		
(-135, 45, 1)	-3,0103 dB		-3,0103 dB
(135, 45, 1)		-3,0103 dB	-3,0103 dB

### 7.3.19 GroupEquatorOrUpper18

#### 7.3.19.1 Channel Layouts in GroupEquatorOrUpper18

Channel layout bitmasks that use GroupEquatorOrUpper18 fold-down coefficients are listed in Table 47.

**Table 47: GroupEquatorOrUpper18 list\_of\_channel\_masks[]**

0x0000A19E	0x0006A19E	0x3000019E	0x3006019E
0x0000A19F	0x0006A19F	0x3000019F	0x3006019F
0x0000A786	0x0006A786	0x30000786	0x30060786
0x0000A787	0x0006A787	0x30000787	0x30060787

#### 7.3.19.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper18

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 48.

**Table 48: GroupEquatorOrUpper18 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	Lsr	Rsr
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]	
(-135, 45,1)	0,0 dB	
(135, 45,1)		0,0 dB

### 7.3.20 GroupEquatorOrUpper19

#### 7.3.20.1 Channel Layouts in GroupEquatorOrUpper19

Channel layout bitmasks that use GroupEquatorOrUpper19 fold-down coefficients are listed in Table 49.

**Table 49: GroupEquatorOrUpper19 list\_of\_channel\_masks[]**

0x0030019E	0x00300786	0x0036019E	0x00360786
0x0030019F	0x00300787	0x0036019F	0x00360787

#### 7.3.20.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper19

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 50.

**Table 50: GroupEquatorOrUpper19 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	L	R	Lsr	Rsr	ExclusiveOr (Lhs,Ltm)	ExclusiveOr (Rhs,Rtm)
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]					
(-45, 45,1)	-3,0103 dB				-3,0103 dB	
(45, 45,1)		-3,0103 dB				-3,0103 dB
(-135, 45,1)			-3,0103 dB		-3,0103 dB	
(135, 45,1)				-3,0103 dB		-3,0103 dB

### 7.3.21 GroupEquatorOrUpper20

#### 7.3.21.1 Channel Layouts in GroupEquatorOrUpper20

Channel layout bitmasks that use GroupEquatorOrUpper20 fold-down coefficients are listed in Table 51.

**Table 51: GroupEquatorOrUpper20 list\_of\_channel\_masks[]**

0x0040419E	0x00404786	0x0046419E	0x00464786
0x0040419F	0x00404787	0x0046419F	0x00464787

### 7.3.21.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper20

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 52.

**Table 52: GroupEquatorOrUpper20 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	L	R	Lsr	Rsr	Ch	Chr
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]					
(-45, 45,1)	-3,0103 dB				-3,0103 dB	
(45, 45,1)		-3,0103 dB			-3,0103 dB	
(-135, 45,1)			-3,0103 dB			-3,0103 dB
(135, 45,1)				-3,0103 dB		-3,0103 dB

### 7.3.22 GroupEquatorOrUpper21

#### 7.3.22.1 Channel Layouts in GroupEquatorOrUpper21

Channel layout bitmasks that use GroupEquatorOrUpper21 fold-down coefficients are listed in Table 53.

**Table 53: GroupEquatorOrUpper21 list\_of\_channel\_masks[]**

0x0040A19E	0x0046A19E	0x3040019E	0x3046019E
0x0040A19F	0x0046A19F	0x3040019F	0x3046019F
0x0040A786	0x0046A786	0x30400786	0x30460786
0x0040A787	0x0046A787	0x30400787	0x30460787

#### 7.3.22.2 Virtual Speakers and their Fold-down Coefficients for GroupEquatorOrUpper21

Since there are no virtual speakers in the equatorial plane, set the corresponding arrays to an empty list.

```
eqtr_functions = [];
eqtr_virt_srcs = [];
```

Fold-down coefficients for virtual speakers in the upper hemisphere are derived from Table 54.

**Table 54: GroupEquatorOrUpper21 Fold-down Coefficients for Virtual Speakers in Upper Hemisphere**

Physical Speakers uphemi_functions	Lsr	Rsr	Chr
Virtual Speakers uphemi_virt_srcs	uphemi_folddown[][]		
(-135, 45,1)	-3,0103 dB		-3,0103 dB
(135, 45,1)		-3,0103 dB	-3,0103 dB

## 7.4 Lower Hemisphere

### 7.4.1 Conditions for Lower Hemisphere Groups

The custom virtual sources for lower hemisphere were separated as there are only five tables shared with multiple groups defined in clause 7.3. For all the tables, the virtual sources in the lower hemisphere is fixed:

$$V_l = [(-45^\circ, -45^\circ, 1), (45^\circ, -45^\circ, 1), (-135^\circ, -45^\circ, 1), (135^\circ, -45^\circ, 1)]$$

The channel layout masks defined in the lower hemisphere groups require that the upper hemisphere channels be masked to 0. The function `GetChannelLayoutMaskForLowerHemisphere()` masks the LFE and upper hemisphere channels to zero.

## 7.4.2 Functions for Virtual Speakers in the Lower Hemisphere

### 7.4.2.1 GetChannelLayoutMaskForLowerHemisphere

`GetChannelLayoutMaskForLowerHemisphere()` converts physical speaker layout into a masked layout for the look up tables in clause 7.4.2.2. The channel masks created by `GetChannelLayoutMaskForLowerHemisphere()` find the virtual sources and corresponding fold-down coefficients according to the bitmask corresponding to their respective lower grouping, found in clauses 7.4.3 to 7.4.7.

```
GetChannelLayoutMaskForLowerHemisphere( physical_speaker_channel_layout_mask )
{
    // Defining masking flags that will be masked to zero for lower hemisphere.
    mask_lfe = LFE1 | LFE2;
    mask_upper_hemisphere = Lh | Rh | Ch | Lhr | Chr | Rhr | Lhs | Rhs | Oh | Ltf | Rtf | Ltr | Rtr;

    // Take negation of both lfe & upper hemisphere masks
    mask_to_zero = (~mask_lfe) | (~mask_upper_hemisphere);

    // Apply conjunction to make lfe's and upper hemisphere channel masks zero.
    channel_layout_mask_for_lower_hemisphere =
        physical_speaker_channel_layout_mask & mask_to_zero;

    // return the resultant channel mask.
    return channel_layout_mask_for_lower_hemisphere;
};
```

### 7.4.2.2 GroupLower

The following structure gives a concrete representation for a group in the lower hemisphere virtual speakers.

```
struct GroupLower
{
    // An array listing all the channel masks supported by the group.
    uint32 list_of_channel_masks[];

    // An array of lower hemisphere virtual speakers used by the group.
    Point lohemi_virt_srcs[];

    // An array of function pointers to determine the lower hemisphere virtual speaker index.
    FunctionP lohemi_functions[];

    // Table of fold-down coefficients for the lower hemisphere virtual speakers.
    GainCoefficient_dB lohemi_folddown[][];

    // Get linear gain vector for virtual sources in lower hemisphere
    GetLinearGainMatrixLoHemi(Point speaker_locations[]);
};

GroupEquatorOrUpper::GetLinearGainMatrixLoHemi(Point speaker_locations[])
{
    out_gain_matrix = [];

    for ( virtual_p_idx : range(lohemi_virt_srcs.size() ) )
    {
        out_gain_matrix[virtual_p_idx] =
            GetLinearGainVectorFromTable(
                lohemi_virt_srcs,
                lohemi_functions,
                virtual_p_idx,
                speaker_locations );
    }
    return out_gain_matrix;
}
```

### 7.4.2.3 list\_of\_all\_groups\_lower

The following list defines all the supported groups for custom virtual speakers in the upper hemisphere and the equatorial plane.

```
list_of_all_groups_lower = {
    GroupLower1,
    GroupLower2,
    GroupLower3,
    GroupLower4,
    GroupLower5
};
```

### 7.4.3 GroupLower1

#### 7.4.3.1 Channel layouts in GroupLower1

Channel layout bitmasks that use GroupLower1 fold-down coefficients are listed in Table 55.

**Table 55: GroupLower1 list\_of\_channel\_masks[]**

0x00000006	0x00001800	0x00060000
------------	------------	------------

#### 7.4.3.2 Virtual Speakers and their Fold-down Coefficients for GroupLower1

Fold-down coefficients for virtual speakers in the lower hemisphere are derived from Table 56.

**Table 56: GroupLower1 Fold-down Coefficients for Virtual Speakers in Lower Hemisphere**

Physical Speakers lohemi_functions[]	ExclusiveOr (L,Lw,Lc)	ExclusiveOr (R,Rw,Rc)
Virtual Speakers lohemi_virt_srcs[]		
(-45, -45,1)	0,0 dB	
(45, -45,1)		0,0 dB
(-135, -45,1)	0,0 dB	
(135, -45,1)		0,0 dB

### 7.4.4 GroupLower2

#### 7.4.4.1 Channel layouts in GroupLower2

Channel layout bitmasks that use GroupLower2 fold-down coefficients are listed in Table 57.

**Table 57: GroupLower2 list\_of\_channel\_masks[]**

0x00000007	0x00001807	0x00060001	0x00060606	0x00061807
0x00000606	0x00001E00	0x00060006	0x00060607	0x00061E00
0x00000607	0x00001E01	0x00060007	0x00061800	0x00061E01
0x00001801	0x00001E06	0x00060600	0x00061801	0x00061E06
0x00001806	0x00001E07	0x00060601	0x00061806	0x00061E07

#### 7.4.4.2 Virtual Speakers and their Fold-down Coefficients for GroupLower2

Fold-down coefficients for virtual speakers in the lower hemisphere are derived from Table 58.

**Table 58: GroupLower2 Fold-down Coefficients for Virtual Speakers in Lower Hemisphere**

Physical Speakers lohemi_functions[]	ClosestAzEq (-40)	ClosestAzEq(40)	LeftLargestEqtr()	RightLargestEqtr()
Virtual Speakers lohemi_virt_srcs[]	lohemi_folddown[][]			
(-45, -45, 1)	0,0 dB			
(45, -45, 1)		0,0 dB		
(-135, -45, 1)			0,0 dB	
(135, -45, 1)				0,0 dB

#### 7.4.5 GroupLower3

##### 7.4.5.1 Channel layouts in GroupLower3

Channel layout bitmasks that use GroupLower3 fold-down coefficients are listed in Table 59.

**Table 59: GroupLower3 list\_of\_channel\_masks[]**

0x00000001E	0x0000181E	0x0006001E	0x0006181E
0x00000001F	0x0000181F	0x0006001F	0x0006181F

##### 7.4.5.2 Virtual Speakers and their Fold-down Coefficients for GroupLower3

Fold-down coefficients for virtual speakers in the lower hemisphere are derived from Table 60.

**Table 60: GroupLower3 Fold-down Coefficients for Virtual Speakers in Lower Hemisphere**

Physical Speakers lohemi_functions[]	L	R	Ls	Rs
Virtual Speakers lohemi_virt_srcs[]	lohemi_folddown[][]			
(-45, -45, 1)	0 dB			
(45, -45, 1)		0 dB		
(-135, -45, 1)			-0,4576 dB	-10 dB
(135, -45, 1)			-10,0 dB	-0,4576 dB

#### 7.4.6 GroupLower4

##### 7.4.6.1 Channel layouts in GroupLower4

Channel layout bitmasks that use GroupLower4 fold-down coefficients are listed in Table 61.

**Table 61: GroupLower4 list\_of\_channel\_masks[]**

0x00000005E	0x0000185E	0x0006005E	0x0006185E
0x00000005F	0x0000185F	0x0006005F	0x0006185F

### 7.4.6.2 Virtual Speakers and their Fold-down Coefficients for GroupLower4

Fold-down coefficients for virtual speakers in the lower hemisphere are derived from Table 62.

**Table 62: GroupLower4 Fold-down Coefficients for Virtual Speakers in Lower Hemisphere**

Physical Speakers lohemi_functions[]	L	R	Ls	Rs	Cs
Virtual Speakers lohemi_virt_srcs[]	lohemi_folddown[][]				
(-45, -45,1)	0 dB				
(45, -45,1)		0 dB			
(-135, -45,1)			-3,0103 dB		-3,0103 dB
(135, -45,1)				-3,0103 dB	-3,0103 dB

### 7.4.7 GroupLower5

#### 7.4.7.1 Channel Layouts in GroupLower5

Channel layout bitmasks that use GroupLower5 fold-down coefficients are listed in Table 63.

**Table 63: GroupLower5 list\_of\_channel\_masks[]**

0x00000019E	0x0000199E	0x0006019E	0x0006199E
0x00000019F	0x0000199F	0x0006019F	0x0006199F
0x0000001DE	0x000019DE	0x000601DE	0x000619DE
0x0000001DF	0x000019DF	0x000601DF	0x000619DF
0x000000786	0x00001F86	0x00060786	0x00061F86
0x000000787	0x00001F87	0x00060787	0x00061F87
0x0000007C6	0x00001FC6	0x000607C6	0x00061FC6
0x0000007C7	0x00001FC7	0x000607C7	0x00061FC7

#### 7.4.7.2 Virtual Speakers and their Fold-down Coefficients for GroupLower5

Fold-down coefficients for virtual speakers in the lower hemisphere are derived from Table 64.

**Table 64: GroupLower5 Fold-down Coefficients for Virtual Speakers in Lower Hemisphere**

Physical Speakers lohemi_functions[]	L	R	Lsr	Rsr
Virtual Speakers lohemi_virt_srcs[]	lohemi_folddown[][]			
(-45, -45,1)	0 dB			
(45, -45,1)		0 dB		
(-135, -45,1)			0 dB	
(135, -45,1)				0 dB

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## History

<b>Document history</b>		
V1.1.1	January 2018	Publication