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3LD – Library for Loudspeaker Layout Design

A Matlab Library for Rendering and Evaluating Periphonic Loudspeaker Layouts.

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Abstract

The generation of loudspeaker layouts for periphonic sound spatialization systems is a non-trivial task, just like the underlying mathematical problem: the homogeneous distribution of a number of points on the surface of a sphere. According approaches have a long tradition in mathematics (platonic solids, geodesic spheres) as well as in physics (minimal energy configurations). However, neither of these fullfills all the requirements of periphonic loudspeaker layouts, such as arbitrary choice of the number of loudspeakers, possibility of psychoacoustical optimization (loudspeaker density as a function of the ear's spatial resolution) and the consideration of practical limitations (forbidden and explicitely demanded loudspeaker positions).

Through a hybrid approach of the strategies mentioned above, it is possible to overcome the limitations of each single method and create a universal tool for the design of three-dimensional loudspeaker layouts. The theoretical background for this has been described in the author's diploma thesis [1]. This project was concerned with the practical implementation of the developed theory and has resulted in the 3LD Library for Loudspeaker Layout Design, which includes features for the generation, visualization and evaluation of periphonic loudspeaker layouts.

1 Periphonic Loudspeaker Layouts

In [1], a detailed description of different criteria regarding the design of 3D loudspeaker layouts has been given, which can be briefly summarized by stating that the design of a periphonic (i.e. 3D) loudspeaker layout has to take into account

- The applied soundfield reconstruction algorithms, e.g. VBAP, Ambisonics, etc.
- The homogeneity of soundfield reconstruction
- The properties of human spatial hearing
- The loudspeaker distribution in the horizontal plane
- The architectural circumstances of a periphonic sound system

In this chapter, we will evaluate different approaches of generating periphonic loudspeaker layouts according to these criteria, starting with the *Platonic solids*, which are optimal configurations regarding the homogeneity of soundfield reproduction.

1.1 Platonic Solids

The five Platonic solids are the only convex polyhedra which are mathematically regular. Figure 1 shows them in the following left-to-right order:

- **Tetrahedron**
- Hexahedron (Cube)
- Octahedron
- Icosahedron
- Dodecahderon

Figure 1: The five Platonic solids

In [2], a definition of regularity regarding Ambisonic soundfield reconstruction has been given, yielding that all Platonic solids are regular in the Ambisonic sense for first and the dodecahedron and icosahedron even for second order systems. Unfortunately, the Platonic solids are not suitable for large-scale periphonic audio reproduction systems, since they provide 20 vertices at the most. We will thus consider the possibility of their *geodesic extension* in the next chapter.

1.2 Geodesic Spheres

By tessellating the facets of a polyhedron and pushing the such created new vertices out to the radius of the original configuration, *geodesic spheres* can be built from the platonic solids or other polyhedra. An example of this iterative process is shown in figure 2. The method of geodesic spheres has been generalized in [1] towards maximum flexibility regarding the choice of the number of loudspeakers in a configuration, resulting in a set of tessellation rules. By applying these rules independently onto different facet shapes and different iterations of the process, we achieve significant freedom in the design of periphonic loudspeaker layouts.

Figure 2: Construction process of a geodesic sphere

1.3 Minimal Energy Configurations

An approach from physics to the distribution of an arbitrary number of points on a sphere are so-called *minimal energy configurations*, which are generated by a random distribution of electrons on a spherical surface: Due to the repulsion forces among the electrons, they will arrange themselves in a natural equilibrum of minimal potential energy after some time. Note that the elctrons are only allowed to move on the surface of a sphere. Figure 3 shows some snapshots of this iterative process, which yields that the homogeneity of the configuration increases with the number of iterations. The obvious advantage of arbitrary numbers of electrons/loudspeakers has to be traded off for a lack of symmetry in the resulting layout.

Figure 3: Electrons distributing themselves towards a minimal energy configuration

2 An Extended Loudspeaker Lay out Design Strategy

In the author's master thesis [1], a new strategy for the design of periphonic loudspeaker layouts has been presented, which is a hybrid approach of the methods discussed so far. It bases on the separation of the design process into two stages, distribution, which is then psychoacoustically refined in the second stage: The spatial resolution of the human ear is best in the horizontal plane and for the front direction, the first of which is dedicated to the construction of a homogeneous loudspeaker whereas elevated and lateral sound sources can not be localized as well. By providing higher loudspeaker densities in areas of better auditory resolution, we can optimize a layout regarding the total number of loudspeakers. It has been suggested in [1] to use the charges of the electrons in a minimal energy algorithm for the implementation of spherical loudspeaker density functions: higher electron charges result in higher repulsion forces among the electrons, and thus in lower loudspeaker densities. A spherical loudspeaker density function can thus be derived as the inverse of a function representing direction-dependent electron charges (which do not exist in nature but are introduced here as a useful concept). However, a nonconstant electron density also means that we cannot choose the initial electron distribution randomly any more, but rather have to use an initial configuration in which the electrons are already to some degree homogeneously distributed over the sphere. The Platonic solids or their geodesic extensions represent suitable intial layouts for this hybrid approach, which can be further extended in order to account for

- Non-spherical layout surfaces, i.e. a spherical radius function
- Gain and delay calibration due to differing loudspeaker distances
- Areas which do not allow for the mounting of loudspeakers
- Forced loudspeaker positions, i.e. 'locked' electrons

he combination of these considerations results in an *extended loudspeaker layout* T *design strategy* shown in figure 4

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Figure 4: Extended Loudspeaker Layout Design Strategy

3 3LD – a Matlab Library for Periphonic Loudspeaker Layout Design

In this chapter, we will present the functions included in the 3LD Library for Loudspeaker Layout Design, including descriptions of their functionality, their input typing "help <functionname>" in the Matlab command line. At the beginning of each section, the presented function is denoted including all of its input and output and output arguments and example applications. These can also be retrieved by arguments. Optional input arguments are denoted in brackets. Note that the cartesian and spherical coordinate system conventions used in all of the functions follow the convention which is also used by native Matlab functions like sph2cart.

3.1 Core Functions

functions, with a range of possible applications clearly exceeding the field of periphonic loudspeaker layout generation: Two core functions have been implemented as useful extensions to native Matlab

spharmonic

functions of degree n and $m*$ sig = -n,-n+1,..,-1,0,1,..,n-1,n, where m is the order and $sig = ±1$ the superscript of a spherical harmonic. Note that the terms 'degree' and 'order' are sometimes used the other way around $-$ for example, $\bm{{\rm n}}$ *rder* of an Ambisonic system. The functions are evaluated for each denotes the *o* $Y = spharmonic(n,az,elev[,norm])$ computes the spherical harmonic element of az and elev. n must be a scalar integer, and az and elev must be arrays of identical size containing azimuth and elevation in radians. norm is an optional argument, specifying different normalizations of the Legendre polynomials, which are used in the calculation of the spherical harmonics. Legal strings are 'unnorm', 'sch' Or 'norm'. The default value is 'unnorm'. spharmonic calls Matlab's legendre with this argument – more information on the normalization options can be found there.

The returned array Y has one more dimension than az and elev. Each element $Y(m^*sign+1,i,j,k,...)$ contains the spherical harmonic function of degree n, order m, and superscript sig, evaluated at $az(i,j,k,...)$, elev(i,j,k,...).

For our spherical coordinate system, the spherical harmonic functions are given as:

```
Y(n,m,sigiaz,elev) =
```

```
sin(m*az) for sig=-1
= P(n, m; sin(elev)) * <
                  \vert cos(m*az) for sig=+1
 _
```
For $m=0$, spharmonic only calculates the term for sig=+1, since the solution for $SIG=-1$ is always zero. $P(n,m;\sin(\text{elev}))$ is the Legendre polynomial of degree n and order m, calculated at $sin(elev)$ by the means of the Matlab function legendre.

Examples:

• spharmonic(2, $0.0:0.1:0.2, 0.3:0.1:0.5$) returns the matrix:


```
Y(:, 1, 2, 3) is the same as
  spharmonic(n,az(1,2,3),elev(1,2,3)).• az = rand(2, 4, 5); elev = rand(2,4,5); n = 2;
  Y = spharmonic(n,az,elev);so that size(Y) is 5x2x4x5 and
```
See also: legendre.

ezspherical

h = ezspherical(fct[,ngrid]) is an easy-to-use plotter for spherical functions. fct must be a function handle with two arguments, the first of which is interpreted as the azimuth, and the second one of which is interpreted as the elevation. ngrid specifies the resolution at which the plot is calculated: for both angles, a function value is calculated at every interval $2*pi/ngrid.$ ezspherical actually plots the absolute value of the radius in each direction. Positive and negative function values can be distinguished by color (red for +1, blue for -1 in the default case). Note that ezspherical uses Matlab's surf, so you can set the properties of the returned graphic handle h as in surf.

Example:

• ezspherical can be used to plot spherical harmonic functions, created by spharmonic (3LD). Use handlespharm (3LD) to create an according function handle. For example, to plot the spherical harmonic function of degree 1, order 1, and superscript -1, try

```
fct = \text{handlespharm('Y(1,-1)')};h = ezspherical(fct);
```
See also: handlespharm (3LD), solospharm (3LD), spharmonic (3LD), surf, ezpolar.

.2 Loudspeaker Layout Generation and Modification 3

Note that all 3LD loudspeaker layout generation functions except minenergyconf return a faces/vertices structure which can be directly plotted using Matlab's patch or 3LD's plot3LD. Faces are always oriented counterclockwise as seen from the origin of a configuration.

platonicsolid

p = platonicsolid(shape[,radius]) generates one of the five convex regular polyhedra, which are also refered to as Platonic solids. Those are the tetrahedron, the hexahedron (cube), the octahedron, the dodecahedron, and the icosahedron. shape is a string specifying which polyhedron to generate: Choices are 'tetrahedron' (or 'tetra'), 'hexahedron' (or 'hex' or 'cube'), 'octahedron' (Or 'oct'), 'dodecahedron' (Or 'dodec'), 'icosahedron' (Or 'ico').

radius refers to the radius of the Platonic solid, i.e. the distance of its vertices to the center of the polyhedron, which is always located at the origin of the coordinate system. If not specified, radius defaults to 1.

p has fields vertices and faces. vertices is a V-by-3 matrix with rows representing the x,y,z coordinates of the V vertices, and faces is a F-by-S matrix tetrahedron, octahedron, and icosahedron have triangular faces (S=3), the cube has rectangular faces (S=4), and the dodecahedron has pentagonal faces (S=5). \rm{p} can with each row listing the row indices of the vertices forming one of the F faces of the polyhedron. S refers to the number of vertices in a face of the polyhedron: The be plotted directly using Matlab's patch or 3LD's plot3LD.

Example:

• Plot an icosahedron:

```
p = platonicsolid('ico'); 
plot3LD(p);
```
See also: geosphere (3LD), minenergyconf (3LD), sphere, ellipsoid, cylinder, patch.

bu cky2

b = bucky2([radius]) generates the vertices/faces structure of a truncated replaces its adjacency matrix with a faces matrix. The output structure b has fields icosahedron, also refered to as 'bucky ball'. This function uses Matlab's bucky, but vertices, which is a V-by-3 array specifying the x,y,z coordinates as returned by bucky, and a F-by-6 f aces array with rows containing the row indices of the vertices forming a facet. Note that the Bucky ball consists of hexagons and pentagons. For the rows in the faces matrix representing a pentagon, the last entry is NaN. The output structure B can be plotted directly using Matlab's patch or 3LD's plot3LD.

See also: bucky.

geosphere

p = geosphere(shape[,freq,radius]) generates geodesic spheres from one of the five platonic solids using 3LD's platonicsolid, or from any other vertices/faces structure at its input. Geodesic spheres are constructed either by adding a vertex in the middle of each facet in a polyhedron and connecting it to every other vertex in the facet (figure 5, left and right pictures), or by subdividing the edges of each facet and connecting the new vertices in a way that depends on the facet's shape (figure 5, mid pictures). The first approach can be applied to arbitrarily shaped facets, while the latter can only be applied to triangles or rectangles. Here, the *frequency* f of the geodesic sphere determines into how many parts each edge is subdivided.

Figure 5: Different tessellations of various facet shapes [1]

sphere, or an existing polyhedron. In the first case, choices are 'tetrahedron' (or 'tetra'), 'hexahedron' Or 'hex' Or 'cube', 'octahedron' Or 'oct', 'dodecahedron' Or 'dodec', 'icosahedron' or 'ico'. In the latter case, shape has to s is a V-by-be a structure with fields vertices and faces. vertice 3 matrix with rows representing the x,y,z coordinates of the V vertices, and faces is a F-by-S matrix with each row listing the row indices of the vertices forming one of shape can be a string specifying a platonic solid from which to build a geodesic

the F faces of the polyhedron. S refers to the number of vertices in a face of the polyhedron.

freq is a matrix which determines the frequencies applied in the tessellation and the number of iterations. The j-th column refers to the j-th iteration. The i-th row refers to the polygons with j minus 2 vertices, i.e. triangles for the first row, rectangles for the second, pentagons for the third, and so on. For example, the element $(4,3)$ of $freq$ specifies the frequency applied to any hexagon in our polyhedron in the third iteration. Possible values for freq are:

- $free = 0$ -> faces are not tessellated
- \cdot freq = 1 -> faces are midpoint-triangulated
- freq > 1 -> triangular faces are triangulated, and rectangular faces are rectangulated at the frequency freq.

Note that since any faces with more than 4 vertices can only be midpointtriangulated, elements >1 in the rows i>3 will be clipped to 1. If $freq$ has less rows up with the entries of the last available row. For example, $\text{freq}=[0; 3]$ means that all all faces with more than 4 vertices will be midpoint-triangulated, since that's our only option. If not specified, freq defaults to 2, i.e. triangles are triangulated at frequency than there are different facet shapes in the polyhedron, the missing rows will be filled triangles will not be modified, all rectangles will be rectangulated at frequency 3, and 2, rectangles rectangulated at frequency 2, and pentagons, etc. are midpointtriangulated.

function for its respective direction. Note that in both cases, the radii of existing vertices will be overwritten if shape is an existing polyhedron. If radius is empty or radius refers to the radius of the geodesic sphere, i.e. the distance of its vertices to the center of the sphere. It can be either a scalar or a handle to a function with two arguments, the first of which is interpreted as azimuth, and the second one as elevation. In the first case, all vertices are set to the same radius specified by the scalar. In the second case, each vertex is set to the radius specified by the radius not specified, the polyhedron will be tessellated, but the new vertices will not be pushed out to a sphere or elsewhere.

p is a structure with fields vertices and faces which provide the same properties as required for the shape input argument. It can be plotted directly using 3LD's plot3LD or Matlab's patch.

Examples:

- $p = geosphere('oct', 2, 1); plot3LD(p);$
- $b = \text{bucky2}; p = \text{geosphere}(b,1,1); \text{plot3LD}(p);$
- $p = geosphere('oct',[2 3]); plot3LD(p);$
- radius = $@(az, elev)$ abs(cos(az) $.*$ cos(elev)) + 2;
- $p = geosphere('oct',[2 2 2], radius); plot3LD(p);$

See also: platonicsolid (3LD), minenergyconf (3LD), sphere, ellipsoid, cylinder, patch

minenergyconf

v = minenergyconf(e[,n,radius,repulsion,density,lock]) simulates the process of electrons distributing themselves over the surface of a sphere until they reach what is refered to as a 'minimal energy configuration', i.e. a natural equilibrum of minimal potential energy. In minenergyconf, this algorithm has been extended to allow for arbitrary surface shapes, a spherical electron density function, scalable repulsion forces, and 'locked' electrons.

e can be a scalar, defining the number of electrons in a new configuration. In this case, the initial positions of the e electrons will be randomly distributed. e can also be an F-by-3 array, representing the x,y,z coordinates of an existing configuration with F electrons, which is then used as the initial layout for further modification.

among all possible electron pairs are calculated, and the electrons are moved to their according new positions. If not specified, n defaults to 1. n represents the number of iterations applied. In each iteration, the repulsion forces

represent a handle to a spherical function, in which case it represents a surface with radius determines the radius of the configuration. If radius is a scalar, the electrons will be distributed on a sphere with that radius. However, radius can also a radius depending on azimuth (first argument) and elevation (second argument). Both angels should be specified in radians. If not specified, radius defaults to 1.

repulsion specifies the power of the repulsion forces among the electrons. If not specified, repulsion defaults to 2, which represents the natural case of repulsion forces between two electrons which are proportional to their inverse square distance. If e.g. repulsion=1, the forces will be proportional to their plain inverse distance.

 $\mathtt{density}$ is a handle to a spherical function representing the electron density as a function of direction (azimuth $=$ first argument, elevation $=$ second argument), in of the inverse densities at their positions. If not specified, $density$ defaults to a constant density of 1 over the entire surface. exactly the same way as RADIUS defines the radius as a spherical function. Higher repulsion forces among the electrons will occur in areas of lower density and vice versa. The repulsion forces among two electrons are then proportional to the product

undersized, the remaining electrons will be considered unlocked. lock is a vector with a length matching the number of electrons in the configuration. A non-zero entry means that the electron with the according row index in the output array V will be 'locked', i.e. it will exercise repulsion forces on the other electrons, but is immune to the forces exercised on itself and will thus remain in its initial position. If not specified, all electrons remain unlocked, i.e. lock is a null vector. If lock is

V is an e-by-3 matrix specifying the resulting x,y,z coordinates of the e electrons.

Example:

• A minimal energy configuration with 50 electrons and 20 ierations:

```
pvertices = minenergyconf(50, 20);
```
See also: platonicsolid (3LD), geosphere (3LD).

rotate_xyz

v = rotate_xyz(coord,axis,angle) rotates the P-by-3 array coord, which represents the x,y,z coordinates of P points, around either $AXIS$ (specified by strings 'x', 'y', or 'z') by an angle given in radians. Arbitrary rotation axes can be achieved by subsequent application of rotate_xyz.

Example:

• Rotate an octahedron 45 degrees around the x axis

```
p = platonicsolid('oct');
```

```
p.vertices = rotate_xyz(p.vertices,'x',pi/4);
```
See also: platonicsolid (3LD), geosphere (3LD), bucky2 (3LD), minenergyconf (3LD).

ma p_to_surface

s = map_to_surface(V[,radius]) maps the N vertices of a polyhedron represented by the N-by-3 array ∇ which contains their x,y,z coordinates - to a surface defined by a function handle radius, which represents the radius as a spherical function. The first argument of the function handle is interpreted as the azimuth, and the second as the elevation. radius can also be a scalar, in which case the vertices are mapped to a sphere of that radius. If r adius is not specified, the vertices are mapped to the unit sphere of radius 1. The N-by-3 output array represents the new x,y,z coordinates of the vertices. Only the radius of the vertices will be affected, whereas their direction is maintained.

Example:

• Create a geodesic sphere and map its vertices to a surface derived from a spherical harmonic function, using 3LD functions. Plot the original polyhedron, the radius function and the new polyhedron.

```
p = geosphere('ico', [2, 2], 1);patch(p,'facecolor',[.1 .7 .3],'facealpha',0.8); 
radius = handlespharm('abs(3*Y(7,-5))+1');
figure; ezspherical(radius,200);
p.vertices = map_to_surface(p.vertices, radius);
figure; patch(p,'facecolor',[.1 .7 .3],'facealpha',0.8);
```
Se e also: handlespharm (3LD), ezspherical (3LD), calibrate_layout (3LD).

3.3 Loudspeaker Driving Signal Generation

am b3d_encoder

B = amb3d_encoder(M,src[,gain,identical,sort]) calculates the weightings of the spherical harmonic components of a 3D Ambisonic soundfield. The N3D encoding convention as specified in [1] is applied.

 M is the Ambisonic order and determines the number of Ambisonic channels $N =$ $(M+1)^2$, which is the number of rows in the field gain of the output structure B.

src specifies the directions of the virtual sound sources. It is an S-by-2 array representing azimuth and elevation in radians.

gain specifies the gains of the virtual sound sources. It can be a scalar, in which source. Additional values will be ignored, and missing values will be set to 1. If not specified, gain defaults to 1 for all sources. case it is applied to all sources, or an S-by-1 array, with different gain factors for each

identical specifies whether the spatialized sound sources are fed by the same audio signal. If identical is not zero, this is assumed to be the case, while independent sources are assumed if identical is 0. This affects the dimensions of the output array B. gain. If not specified, identical defaults to 0.

sort is a string which affects the order of the rows in the output array. For 'spharmonic' (or 'sph' or 's'), the order of rows matches the one of 3LD's spharmonic, while for 'ambisonic' (or 'amb' or 'a'), the Ambisonic channels are sorted according to a convention used in [1]. If not specified, sort defaults to 'spharmonic'.

The output structure B has two fields. B.gain is an N-by-S matrix if identical=0 and an N-by-1 matrix if identical=1 and represents the weightings of the N spherical harmonic components. In the latter case, all sound sources are fed by the same signal and their weightings are superponed. B sort is equivalent to the sort

input argument. It is added to the output structure for later decoding with 3LD's amb3d_decoder.

Example:

• Encode a front and a top source at third order source_position = $[0 0:0 \text{ pi}/2]$; source_gain = $[1 0.5]$; B = amb3d_encoder(3,source_position,source_gain)

See also: amb3d decoder (3LD), amb3d regularity (3LD), vbp (3LD), spharmonic (3LD).

am b3d_decoder

g = amb3d_decoder(B,spk[,M,method,flavor]) derives the loudspeaker gains for an array of L speakers from the Ambisonic channel gains of a 3D encoded soundfield.

B, which typically represents the output of 3LD's amb3d_encoder, has to be a structure with fields gain, and sort. gain is an N-by-S array, with rows representing the N Ambisonic channels of S sound sources represented in the columns. sort determines the order in which the Ambisonic channels appear. Check the documentation of amb3d_encoder for a description.

spk specifies the directions of the loudspeakers. It must be an L-by-2 array, representing azimuth and elevation in radians. Note that amb3d_decoder calls amb3d_encoder for re-encoding the loudspeaker layout. Thus, the N3D encoding convention [2] is applied in this process

M is the Ambisonic order at which the decoder operates. If not specified, it is derived from the encoded input material B. Otherwise, the order of the input material can only be overridden with smaller values.

method specifies the decoding method. Legal strings are 'projection' (or 'proj') and 'pseudoinverse' (or 'pinv').

flavor specifies the decoder flavor. Legal strings are 'basic' and 'inphase'.

reproduce one of the S independent virtual sound sources, which are represented in the rows. Note that the function does by no means normalize the output gains. It is g is an L-by-S matrix. Each column represents the loudspeaker gains required to your responsibility to avoid clipping.

Example:

• Decode the output of the example in the AMB3D_ENCODER documentation to an icosahedron loudspeaker layout at second Ambisonic order.

```
p = platonicsolid('ico');
```

```
[az \text{ elev}] = \text{cart3sph}(p.\text{vertices});
```
g = amb3d_decoder(B,[az elev],2,'pinv','inphase')

See also: amb3d_encoder (3LD), amb3d_regularity (3LD), vbp (3LD), spharmonic (3LD).

am b3d_regularity

 $[regularity, condnum, C] = amb3d_regularity(M,spk)$ returns a string regularity, which specifies whether a 3D layout of L loudspeakers is 'regular', 'semiregular' or 'irregular' in the Ambisonic sense for a specific Ambisonic order M, as defined on page 176 in [1]. It also returns the condition number $\verb|condnum|$ of the re-encoding matrix C, which can be built from the loudspeaker position formation and is also a regularity criterion according to [3]. The third output in argument c is the matrix (C*C')/L, which is used to evaluate the regularity. The layout is regular if this matrix is the unity matrix, and semi-regular if it is a diagonal matrix. Note that slight variations have been allowed in order to account for numerical inaccuracies. spk is an L-by-2 matrix containing the azimuths (first column) and elevations (second column) of the L loudspeakers. Note that as usual, the N3D encoding convention [1] is applied in evaluating the regularity and condition number of the layout.

Example:

• A dodecahedron is regular for second order Ambisonic

```
p = platonicsolid('dodec'); 
[az \text{ elev}] = \text{cart3sph}(p.\text{vertices});
amb3d_regularity(2,[az elev])
```
See also: amb3d_encoder (3LD), amb3d_decoder (3LD).

vbp

g = vbp(src,spk,group[,type,gain,identical]) calculates the gains of an array of L loudspeakers due to S vector-base panned virtual sound sources. 2D or 3D VBP can be applied.

 src specifies the directions of the virtual sound sources. For 2D VBP, it is an S-by-1 array representing the azimuths, whereas for 3D VBAP VBAP, it is an S-by-2 array representing azimuths and elevations. All angles have to be specified in radians.

spk specifies the directions of the loudspeakers. It can be an L-by-1 (2D VBP) or an L-by-2 (3D VBP) array, and is interpreted in the same way as SRC.

group specifies pairs (2D VBP) or triples (3D VBP) of loudspeakers. and thus is an R-by-2 or R-by-3 matrix, where R is the number of pairs/ triples. The entries of group represent the row indices of the respective loudspeakers in spk.

 $type$ is a string that specifies whether vector base amplitude panning $[4]$ or vector or 'a'. For the second case, 'vbip' or 'i' are legal. If not specified, this argument defaults to 'vbap'. base intensity panning [5] is applied. For the first case, the string should be 'vbap'

case it is applied to all sources, or an S-by-1 array, if different gain factors for each source are to be applied. Additional values will be ignored, and missing values will be gain specifies the gains of the virtual sound sources. It can be a scalar, in which set to 1. If not specified, gain defaults to 1.

identical specifies whether the spatialized sound sources are fed by the same audio signal. If identical is not zero, this is assumed to be the case, while independent sources are assumed if $\mathtt{identical}$ is 0. This affects the dimensions of the output array G. If not specified, identical defaults to 0.

the sound sources are fed by the same audio signal (i.e. ${\tt identical=1}$) or an L-by-S matrix if not (i.e. identical=0). In the latter case, the columns of g represent the The output array g represents the gains of the loudspeakers. It is an L-by-1 matrix if gain factors for the S independent sound sources. Note that the function does by no means normalize the output gain factors. It is your responsibility to avoid clipping.

Example:

• Reproduce a front source on an octahedron loudspeaker layout

```
source_position = [0 0]; 
p = platonicsolid('oct'); 
[az \text{ elev}] = cart3sph(p, vertices);g = vbp(source_position,[az elev],p.faces,'vbip')
```
See also: amb3d_encoder (3LD), amb3d_decoder (3LD), amb3d_regularity (3LD).

ca librate_layout

 $[g,d] = calibrate_layout(sppk[,c])$ calculates the calibration gains g and delays d for an array of loudspeaker with varying radii. s pk is a L-by-3 array representing the x,y,z coordinates of the L loudspeakers in the array. The same sound, which defaults to 343.3 meters per second (air at 20°C and at sea level) if not specified. cartesian coordinate system as in Matlab's SPH2CART is applied. c is the speed of

 g is a vector of length L, representing the gain factors of all loudspeakers in the array due to the 1/r law of sound pressure amplitude decay. The speaker with the greatest radius will be assigned a gain factor of 1, whereas the gains of the closer speakers will be attenuated to factors < 1.

d is a vector of length L, representing the delay factors of all loudspeakers in the array due to the finite speed of sound C in seconds. The speaker with the greatest radius will be assigned a gain factor of 0.

Note that loudspeaker array calibration results in the virtual sound sources moving at a distance which is equal to the maximum radius of a loudspeaker in the array, i.e. the distance of the loudspeaker which the layout is normalized too.

Example:

• Create a platonic solid using a 3LD function and move one of its vertices to a greater radius. Calculate the calibration gains and delays for loudspeakers positioned at the vertices of the solid.

```
p = platonicsolid('tetra'); 
p.vertices(1,:) = [0.8, 0.8, 0.8];
p.vertices 
[g,d] = calibrate_layout(p.vertices)
```
See also: map_to_surface (3LD).

3.4 Soundfield Rendering and Evaluation

so undfielder

```
[p,v,VV,uV] = soundfielder(src,freq,type,X,Y,Z,time 
..[,gain,sum,dir,T]) 
.
```
calculates the complex sound pressure, velocity, velocity vector, and u velocity of soundfields in air, created by S monochromatic sound sources with different radiation haracteristics. The positions of the sources and sinks of the field can be arbitrarily c defined.

 src an S-by-3 array containing the x,y,z position information of the S sound sources.

freq is a scalar or a vector with lenght S, either refering to an identical frequency of all monochromatic sound sources, or specifying the frequency of each source independently.

type is a string which determines the radiation characteristics of the sound sources. 'p'. If type is a single string, all sound sources will be of that type. If can also be a Possible choices are 'spherical' or 'sph' or 's', and 'plane' or 'pl' or cell array of S strings, specifying the type for each of the sources independently.

 x , y , and z are arrays specifying the x , y , and coordinates of the sinks, i.e. the measuring points of the soundfield. Use the output of Matlab's meshgrid for these the three output arguments v , vv , and u v . In all cases, the arrays have to be of identical size, and an equal number of dimensions is used in the output arrays to arrays, i.e. 3D arrays for cubic, 2D arrays for plane, or vectors for line sink definition. soundfielder uses the number of dimensions for gradient evaluation of the output vector fields, thus you have to follow the described scheme to get useful output for represent them. The only exception is given when x and y are vectors and z is a scalar, in which case soundfielder does the meshgriding itself such that all combinations of x and y at constant height z are calculated, i.e. the soundfield on a horizontal plane. Accordingly, two dimensions are used in the output arrays to represent the sinks.

time is a vector of the points in time at which the soundfield is rendered, which have to be specified in seconds.

rendered, or a vector of length S, representing time-constant gains for each source, or an S-by-N array, where the element (i,j) represents the gain of the i-th sound amplitude as well as the phase of a wavefront. gain determines the gain factors of the S sound sources. It can either be a scalar, refering to identical gains for all sources at each point of time for which the field is source at the j-th point in time. Note that gain can be complex, specifying the

sum is a scalar determining whether the soundfields created by the S sources are superponed in the output arrays P and V , which is the case if sum is a non-negative scalar.

dir specifies the direction of the wavefronts. Non-negative numbers refer to the share the same direction or not. incoming wavefront, while negative numbers specify an outgoing wavefront, dir can either be a scalar or a vector of length S, depending on whether all S sound sources

T is the time-invariant and homogeneous temperature at which the soundwaves propagate. It is specified in Kelvin and defaults to 273.15 Kelvin (0°C) if not specified.

 $_{\rm P}$ is the complex sound pressure field. The first dimensions of $_{\rm P}$ refer to the which the field is rendered. If sum = 0, an additional dimension refers to the fields of dimensions of X, Y, Z (unless those were specified as two vectors and a scalar -> see X, Y, Z), followed by another dimension representing the various points of time for the different sound sources. Thus, generally $size(p) =$ $[size(X),lenqht(time),size(src,1)].$ However, any singleton dimension will be removed, e.g. if you calculate the soundfields caused by three sources in a cubic sink grid at a single point of time, $size(p) = [size(X), 3]$. If you additionally superpone the fields created by the sources, $size(p) = size(X)$.

vector components of the gradient field and thus has as many elements as there are non-singleton dimensions in x, y, z, i.e. three if the sink data is cubic, two if it is \rm{v} is the complex sound velocity field as discussed in [6]. As in \rm{p} , its first dimensions refer to the dimensions of X , Y , Z . The first dimension after these represents the plane, and one if it represents a line. The next dimension represents the elements in time, and if sum~=0, another dimension refers to the velocity fields due to the different sources of the field. Thus, generally $size(V)$ = [size(X),numNonSingletonSinkDims,length(TIME),size(SRC,1)], but as in p, all singleton dimensions are removed, so if you calculate the superponed field of S sources along a line of 10 points in space for 17 points in time, $size(V) =$ [10 17].

VV is the complex velocity vector field, as defined in [7]. It is an array with the same size and properties as v. Its real part is associated with the perception of direction in a soundfield, and its imaginary part is often refered to in the literature as 'phasiness'. As in p, all singleton dimensions will be removed.

properties as V . Its real part is refered to in the literature as 'active velocity', and is associated with the perception of direction in a soundfield. Its imaginary part is UV is the complex u velocity as defined in [6]. It is an array with the same size and refered to as 'reactive velocity' and does not relate to sound energy transport. As in P, all singleton dimensions will be removed.

See also: pressure_errors (3LD), direction_deviation (3LD), gradient, meshgrid.

direction_deviation

direction_deviation(refdir, synthdir) computes errors among the directions of two vector fields. \texttt{refdir} and $\texttt{synthdir}$ are two arrays of equal size, representing the vector fields which indicate the directions of a complex dirdev reference pressure field and a synthesized field. Direction indicators are for example the real part of the complex velocity or the real part of the u velocity as calculated by 3LD's soundfielder.

the vector field. This dimension will be missing in the output array direction_deviation: since the direction deviation is specified as a scalar in dim specifies the dimension which represents the x,y,and possibly z components of radians at each point of the field, the dimensions representing the vector field components becomes singleton and is removed by the function.

See also: soundfielder (3LD), pressure_errors (3LD).

pressure_errors

[pe2, ae] = pressure_errors(ref, synth) computes errors among two complex sound pressure fields ref and synth, where the first represents the original soundfield, and the latter represents the reconstructed soundfield. The size of the input arrays is arbitrary, but has to match. The error is calculated as a scalar field with the size of the two input fields.

e2 is the 'squared sound pressure error', i.e. the squared difference of the two p fields. Note that it is complex. ae is the 'sound pressure amplitude error', which is the absolute value of the difference of the two fields and is real.

See also: soundfielder (3LD), direction_deviation (3LD).

.5 Helper Functions 3

olospharm s

Y = solospharm(n,mTimesSig,az,elev[,norm]) computes the spherical abs(mTimesSig(i)) <= n. az and elev must be arrays of identical size harmonic functions of degree n, order m, and superscript $sig = \pm 1$. The functions are evaluated for each element of az and elev. n must be a scalar integer. mTimesSig must be a vector with each element i fullfilling the condition containing, the azimuth and elevation arguments in radians. norm is an optional argument, specifying different normalizations of the Legendre polynomials, which are included in the spherical harmonics. Legal terms are 'unnorm','sch' or 'norm', and default is 'unnorm'. Y returns the values of the spherical harmonic functions for each element in the mTimesSig vector and each pair az, elev. The first dimension of Y refers to the different spherical harmonic functions, whereas the other dimensions refer to those of the input arrays az and elev. If mTimesSig is a scalar, the first (singleton) dimension is removed, and γ has the same size as az and $e \, \text{lev}$.

solospharm calls spharmonic (3LD), which calls legendre with the norm argument; more information on the normalization options can be found in legendre.

spharmonic always returns the spherical harmonic functions of all orders and both superscripts for the specified degree n, i.e. from m TimesSig = $-n:n$. To be able to access single harmonic function of a certain order, solospharm has been introduced. Note that it is inefficient, since solospharm simply throws away the functions returned by spharmonic which have not been requested. However, spharmonic uses the native Matlab function legendre, which does not return functions of a single order either.

See also: spharmonic (3LD), legendre.

andlespharm h

h = handlespharm(string) returns a handle to a combination of spherical harmonic functions. string is a string defining this combination using terms 'Y(n, $m*$ sig)', where n refers to the degree, m to the order, and sig = ± 1 to the superscript of the spherical harmonic function. Note that $abs(m*sig) \leq n$. The output of handlespharm can be plotted directly with ezspherical (3LD). Note that handlespharm the spherical harmonics in handlespharm are always Schmidtseminormalized.

handlespharm calls solospharm (3LD), which calls spharmonic (3LD), which calls Matlab's legendre. Refer to those functions for more information.

Examples:

• Plot the sum of the spherical harmonic of degree 3, order 2, and superscript +1, and the absolute value of the function of degree 7, order 1, and superscript -1:

```
h = \text{handlespharm}('Y(3,2) + abs(Y(7,-1))');
```
ezspherical(h);

• Butterfly demo

```
h = handlespharm('(Y(4,3)*Y(5,5)) / abs(Y(1,0)+2)');
ezspherical(h);
```
See also: spharmonic (3LD), solospharm (3LD), ezspherical (3LD), legendre.

cart3sph

 $[az, elev, r] = cart3sph(xyz)$ This function is identical to Matlab's cart2sph, but takes a single N-by-3 matrix as an input argument, rather than three separate arrays. The columns of xyz represent the x, y, and z components of N different points. The function returns the az (azimuth), elev (elevation), and r (radius) components as three vectors. This is convenient to evaluate the spherical coordinates of a vertices array as returned by platonicsolid, geosphere, or bucky2 (all 3LD) without prior separation into x,y,z vectors.

See also: sph3cart (3LD).

sph3cart

 $xyz = sph3cart(az, elev, r)$. This function is identical to Matlab's $sph2cart$, but returns a single N-by-3 matrix as an input argument, rather than three separate arrays. The columns of xyz represent the x, y, and z components of the N different points. The azimuth, elavtion and radius components have to be provided as three independent input arguments. This function is convenient for converting the spherical coordinates of a vertices array back to their cartesian representation after edits like radius mapping.

See also: cart3sph (3LD).

eg2rad d

ad = deg2rad(deg) r

Convert an arbitrary numeric input array \deg from degrees to radians.

See also: rad2deg (3LD).

rad2deg

 $deg = rad2deg(rad)$

Convert an arbitrary numeric input array rad from radians to degrees.

See also: deg2rad (3LD).

plot3LD

h = plot3LD(thing[,lock]) is a straightforward-to-use plotter for vertices/faces structures representing loudspeaker layouts and spherical functions representing radius or loudspeaker density functions.

In the case of loudspeaker layouts, thing is a structure with fields 'vertices' and 'faces'. Additionally, you can specify the lock status of the speakers (used also in minenergyconf), which will be represented with the color of the loudspeaker index in the plot: IDs of unlocked speakers are black, whereas locked loudspeaker indices appear red. The plotting is done by Matlab's patch function. 3LD functions which return a structure that can be directly plotted with plot3LD are platonicsolid, bucky2, and geosphere.

For plotting spherical functions, thing is a function handle depending on two variables, the first of which is interpreted as the azimuth and the second one of which as the elevation. The plotting is done by 3LD's ezspherical. The output of 3LD's handlespharm can be plotted directly using plot3LD.

See also: ezspherical (3LD), platonicsolid (3LD), bucky2 (3LD), geosphere (3LD), patch.

.6 Demo Scripts 3

The following demo scripts are included in the 3LD distribution:

- demo_density
- demo_minenergy •
- demo_geosphere

• demo_soundfielder

and can serve as useful starting points for exploring the library.

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