

Lumped Loudspeaker Driver

This is a model of a moving-coil loudspeaker where a lumped parameter analogy represents the behavior of the electrical and mechanical speaker components. This lumped model is coupled to a 2D axisymmetric pressure acoustics model describing the surrounding air domain. The coupling is performed using the built-in Interior Lumped Speaker Boundary feature. The model is solved with two different configurations, first with an open backvolume and then with a closed back-volume. The main part of the analysis focuses on the open back-volume configuration.

Electric circuit representations of transducers are well known and widely used. In the loudspeaker industry such models have been employed for a long time and with great success. The parameters that characterize the low-frequency performance of a loudspeaker are commonly known as the Thiele-Small or the small-signal parameters. It is common to use these parameters to design and simulate a desired speaker performance in terms of, for example, the on-axis speaker response, the electrical impedance, and the diaphragm velocity.

In the low-frequency regime, the motion of the speaker cone can be approximated by the motion of a rigid piston in an infinite baffle. Analytical expressions exist for the sound field radiated from a piston. In the model at hand, the simple piston geometry is replaced by a more realistic speaker cone shape, and the sound field is solved using the finite element method. Of course, this is still an approximation as the motion of the speaker cone is still assumed to be rigid; however, the spatial response of the speaker is more realistic. The methodology of lumping certain parts of a complex system is a general and powerful approach that can be applied to other systems. A first extension of the current model can be to only lump the electrical components and model the mechanical components fully.

The output from the model includes, among many things, the total electric impedance, the on-axis sound pressure level at a nominal driving voltage, and the mean speaker cone velocity. The results are compared with an analytical solution based on the flat piston approximation. Finally, it is also shown how to create a so-called directivity plot of the speaker.

Note: This application requires the Acoustics Module and the AC/DC Module.

A schematic representation of a moving coil loudspeaker is given in Figure 1. The figure shows a cross section of a loudspeaker. The speaker driver is placed in an infinite baffle with free space in front and on the back of the speaker. This will be referred to as the open backvolume configuration. In the closed back-volume configuration, a closed box is located behind the driver (see Figure 3). The speaker cone consists of the outer suspension, the diaphragm, and a dust cap (not marked in the figure). The mechanical and electrical components of the speaker that are lumped are visualized inside the dotted box. On the electrical side it includes the voice-coil and magnetic system (permanent magnet and pole pieces), and on the mechanical side it includes the moving mass of the voice coil and speaker cone, the spring effect of the spider and outer suspension, as well as possible losses due to damping in these suspensions. In the discussion below, the lumped model of the driver itself is not concerned with the open or closed configuration of the back-volume. The effect of where the driver is located or how it is mounted in a system is modeled in the finite element domain.

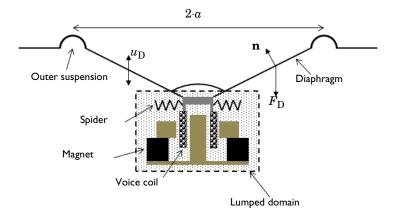


Figure 1: Schematic representation of a moving coil speaker unit.

ELECTROACOUSTIC ANALOGY

In an electroacoustic analogy, mechanical and acoustic physical properties such as force, velocity, pressure, and volume velocity are equated to voltages and currents in an analogous electrical circuit. In this model, the impedance analogy is used for the mechanical components only. This means that the current in the circuit represents the velocity of a moving part (SI unit: m/s) and the voltage represents a force (SI unit: N). In contrast, the acoustics is not represented by a circuit in this model. In general, however,

in an acoustic analogous circuit, a voltage represents pressure (SI unit: Pa) and a current represents volume velocity (SI unit: m³/s).

The passive elements in a circuit — that is, resistors, inductors, and capacitors — represent different processes, namely resistance, mass movement, and compliance, respectively. The components and their analogs are listed in Table 1 (see also Ref. 1 for further details).

TABLE 1: PASSIVE ELEMENTS IN MECHANICAL AND ACOUSTIC ANALOGIES.

ELECTRICAL COMPONENT	MECHANICAL (IMPEDANCE ANALOGY)	ACOUSTIC
ResistorSI unit: Ω	 Mechanical resistance SI unit: N·s/m Losses due to friction, as in a car suspension. 	 Acoustic resistance SI unit: kg/(m⁴·s) Dissipative losses due to viscosity and thermal conduction in the fluid.
• Inductor • SI unit: H	 Mass SI unit: kg Inertial force, acceleration of a mass. 	 Acoustic mass SI unit: kg/m⁴ Internal force of a volume of air that is accelerated but not compressed.
Capacitor SI unit: F	 Compliance SI unit: m/N The inverse of the spring constant. 	 Acoustic compliance SI unit: m⁵/N Compressibility effect of a volume of air that is not accelerated.

The active components in a circuit represent sources; they are external forces (pressures) or applied velocities (volume velocities). The sources are also used for couplings between the electrical, mechanical, and acoustic domains.

Note: To make the units fit in COMSOL, it is necessary to make unit transforms to fit the electrical units. When, for example, inserting a capacitor representing a mechanical compliance, C_{MS} , in the physics interface type: $C_{MS}[F*N/m]$

THE ANALOGOUS CIRCUIT

The analogous circuit for the electrical and mechanical parts of the system sketched in Figure 1 is shown in Figure 2. The upper figure represents the voice coil electrical system and the lower figure the mechanical analogue of the speaker cone, suspensions, and mass of the voice coil. In both figures, the node numbers are also shown — they are very useful when setting up the circuit model in COMSOL.

In Figure 2 (top) the external voltage source is denoted by V_0 and the generator output resistance is $R_{\rm g}$, in this model $R_{\rm g}$ = 0 Ω . The voice coil resistance is $R_{\rm E}$, and the voice coil inductance is $L_{\rm E}(\omega)$, which is frequency dependent. The losses in the magnetic circuit are modeled through the frequency dependent resistance $R'_{\rm E}(\omega)$. The current controlled voltage source $BL\cdot u_{\rm D}$, represents the back induced electromagnetic voltage generated when the voice coil (of length L) moves with velocity $u_{\rm D}$ in the magnetic field B. Here BL is the product of the magnetic field strength and the voice coil length L (see also Ref. 2 on how this can be modeled). In the electrical circuit, the current is denoted $i_{\rm C}$.

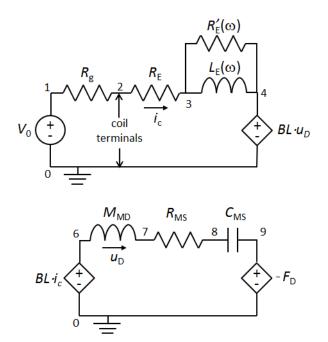


Figure 2: Analogous circuits for the electrical (top) and mechanical (bottom) properties of the speaker driver.

In the mechanical system given in Figure 2 (bottom), the current in the circuit is the velocity of the voice coil and speaker cone in the axial direction (see Figure 1). The force acting on the diaphragm is given by $BL \cdot i_c$. This is the Lorentz force on a voice coil of length L with current i_c , where B is the magnetic flux density. The force acts on a system of mass $M_{\rm MD}$ (voice coil and diaphragm assembly). The resistance $R_{\rm MS}$ models the

damping and $C_{
m MS}$ the mechanical compliance in the speaker suspensions (both spider and outer suspension). Finally, the voltage source $-F_D$ represents the pressure force acting on the speaker diaphragm (in the axial direction). Notice the minus sign that indicates that the force acts against the movement of the diaphragm (see Ref. 1). The force is given by

$$F_D = \int (\Delta p \cdot n_z) dA \tag{1}$$

where Δp is the pressure drop across the diaphragm and n_z is the axial component of the surface normal \mathbf{n} (see Figure 1). This expression gives the couplings from the acoustic finite element model to the lumped circuit model. On the other hand, the coupling from the circuit model to the finite element model comes from specifying the velocity \mathbf{v} on the surface of the diaphragm, which is given by

$$\mathbf{v} = u_{\mathbf{D}} \mathbf{e}_{\mathbf{z}} \tag{2}$$

where $\mathbf{e}_{\mathbf{z}}$ is the unit vector in the axial direction (see Figure 3) and $u_{\mathbf{D}}$ is the current in the mechanical circuit analog. The coupling introduced through Equation 1 and Equation 2 is automatically performed when using the built-in Interior Lumped Speaker Boundary feature (when modeling the air on both sides) or the Lumped Speaker Boundary features. The first is used when setting up the model. Dedicated postprocessing variables can be found in the plot menu for the diaphragm velocity, axial pressure, radiated power and more.

Note: A version of this model using the *Lumped Mechanical System* interface also exists as a tutorial: Lumped Loudspeaker Driver Using a Lumped Mechanical System. In that model, the mechanical part of the system uses the mobility analogy where the present model is based entirely on the impedance analogy. The model additionally requires the Multibody Dynamics Module.

SMALL-SIGNAL PARAMETERS

The fundamental small-signal parameters of the system (or Thiele-Small parameters) are the physical parameters of the loudspeaker driver. They are all constants, as they are given in the low-frequency limit of the model. See Table 2 below.

TABLE 2: FUNDAMENTAL SMALL-SIGNAL PARAMETERS.

SYMBOL	VALUE	DESCRIPTION
$M_{ m MD}$	33.4 g	Moving mass (voice coil and diaphragm)
$C_{ m MS}$	1.18·10 ⁻³ m/N	Suspension compliance
$R_{ m MS}$	1.85 Ns/m	Suspension mechanical losses (damping)
$L_{ m E}$	6.89 mH	Voice coil inductance
$R_{ m E}$	7 Ω	DC resistance of voice coil
BL	11.4 T·m	Force factor
$S_{ m D}$	$a^2\pi$	Driver equivalent area
a	I2 cm	Piston radius of driver (equivalent)

In Table 2 the radius α is the piston radius of the driver. It is typically taken as half the diaphragm aperture diameter, measured half-way into the outer suspension.

Note: The constants used in this model are all taken from Example 22 in Ref. 1.

Other small-signal parameters may be determined on basis of the fundamental small-signal parameters. Actually, these parameters are the ones measured when characterizing a driver; the fundamental parameters are inferred hereafter. The measured small-signal parameters are given in Table 3. In the table, the speed of sound is denoted c_0 and the density of air is ρ_0 . The values of these parameters are determined in the model using the Parameters feature under Global Definitions.

TABLE 3: SMALL-SIGNAL PARAMETERS (MEASURED).

Symbol	Expression	Description
$M_{ m MS}$	$M_{ m MD} + 2 S_{ m D}^2 rac{8 ho_0}{3 \pi^2 a}$	Moving mass including acoustic load (low frequency approximation).
$F_{ m s}$	$rac{1}{2\pi\sqrt{C_{ m MS}M_{ m MS}}}$	Fundamental resonant frequency.

TABLE 3: SMALL-SIGNAL PARAMETERS (MEASURED).

Symbol	Expression	Description
$Q_{ m ES}$	$\frac{2\pi F_{\mathrm{s}} M_{\mathrm{MS}} R_{\mathrm{E}}}{\left(BL\right)^2}$	Electrical Q factor at F_{S}
$Q_{ m MS}$	$rac{2\pi F_{ m s} {M}_{ m MS}}{R_{ m MS}}$	Mechanical Q factor at $F_{ m s}$
$Q_{ m TS}$	$\frac{Q_{\rm MS}Q_{\rm ES}}{Q_{\rm MS}+Q_{\rm ES}}$	Total Q factor at $F_{ m s}$
$V_{ m AS}$	$\rho_0 c_0^2 S_{\mathrm{D}}^2 C_{\mathrm{MS}}$	Equivalent volume compliance (air volume having the same compliance as the suspension).
η_0	$\frac{4\pi^2 F_{_{\rm S}}^3 V_{{\rm AS}}}{c_0^3 Q_{{\rm ES}}}$	Reference efficiency of the driver.

OTHER PARAMETERS

The model also uses other parameters than the small-signal parameters presented above. Two of them are the frequency dependent voice coil inductance $L_{\rm E}(\omega)$ and the resistance $R'_{\rm E}(\omega)$ associated with the losses in the magnetic system. A model for this is given in Ref. 1, for the higher audio frequencies, defining

$$L_{\rm E}(\omega) = \left[\frac{L_{\rm E}}{\sin\left(n_{\rm e}\frac{\pi}{2}\right)}\right] \omega^{(n_{\rm e}-1)} \qquad R'_{\rm E}(\omega) = \left[\frac{L_{\rm E}}{\cos\left(n_{\rm e}\frac{\pi}{2}\right)}\right] \omega^{n_{\rm e}}$$

where n_e is the so-called voice coil loss factor. For $n_e = 1$ and for in the low frequency limit the loss-less behavior is recovered where $R'_{\rm E}$ is an open circuit and $L_{\rm E}$ is constant. In this model $n_e = 0.7$.

Another expression used in the model is the acoustic radiated power which is given by

$$P_{\rm AR} = \int (\mathbf{n} \cdot \mathbf{I}) dA \tag{3}$$

where **I** is the intensity vector. A variable is predefined by the lumped speaker feature and called acpr.ilsb1.P front (front side of the speaker) and acpr.ilsb1.P front (back side of the speaker). The integral is taken over the front and back side of the speaker diaphragm, respectively. The electric input power (RMS value) is defined as

$$P_{\rm E} = 0.5 \operatorname{Re}(V_0 \cdot i_{\rm c}^*) \tag{4}$$

where * is the complex conjugate operator, and, finally, the speaker efficiency comparing the input electric power to the radiated acoustic power is

$$\eta = \frac{P_{AR}}{P_{E}}$$

A number of the analytical results derived in Ref. 1, which are based on the piston approximation, are also used for comparison with the current more realistic speaker cone shape. They are the low-frequency and high-frequency approximations to the speaker velocity, the on-axis pressure at 1 m in front of the piston, and the acoustic radiated power from a piston in an infinite baffle. Expressions for these are given in the reference and under the variable node Component I>Definitions>Analytical approximations in the model.

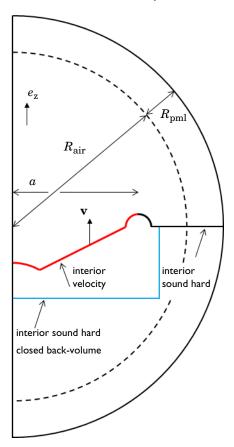


Figure 3: Computational domain and boundary conditions.

FINITE ELEMENT DOMAIN

The computational domain where the pressure acoustics model is solved is sketched in Figure 3. It represents the speaker cone, dust cap, and outer suspension, in an infinite baffle in a 2D axisymmetric model. On the speaker (red line), the Interior Lumped Speaker Boundary feature is applied and the rest of the baffle is an Interior Sound Hard Boundary (Wall) condition. In the closed back-volume configuration, the blue line is also modeled an interior sound hard wall. In the open configuration, no condition is applied to that boundary (it is transparent). The domain is truncated with a perfectly matched layer (PML) to mimic an infinite open domain. Note that the interior sound hard boundary of the baffle should be applied inside the PML. This is to avoid erroneous energy leaks between the two sides of the infinite baffle. The air domain has a radius of $R_{\rm air}$ while the thickness of the PML layer is $R_{\rm pml}$. For more information about PMLs in acoustics, see the Modeling with the Pressure Acoustics Branch (FEM-Based Interfaces) section in the Pressure Acoustics Module User's Guide.

Results and Discussion

In this section, the results of the open back-volume configuration are first discussed in detail. Then selected results are presented for the closed back-volume configuration.

OPEN BACK-VOLUME CONFIGURATION

The generated pressure field is shown in Figure 4 for 1 kHz and 5 kHz. This plot shows the directive characteristic of the speaker cone at increasing frequencies; this nature is discussed more at the end of this section when discussing the directivity plot Figure 11.

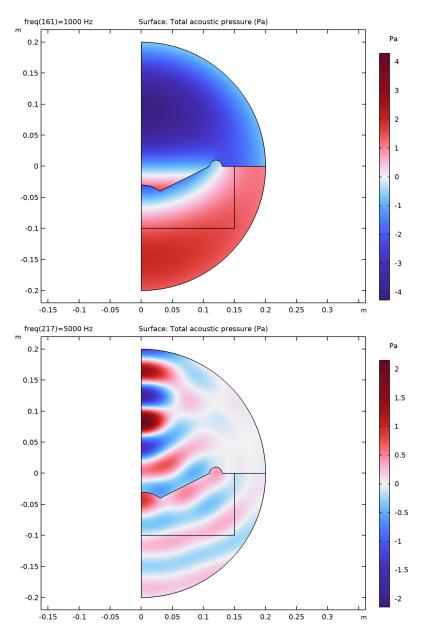


Figure 4: Acoustic pressure for a frequency of 1 kHz (top) and 5 kHz (bottom).

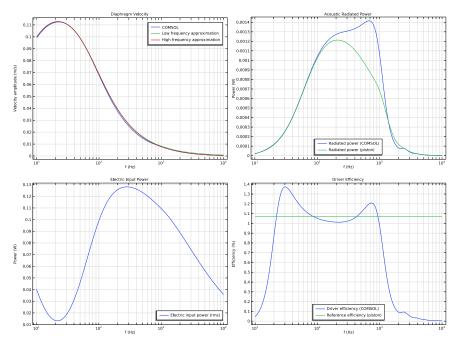


Figure 5: The frequency-dependent diaphragm velocity amplitude (upper left), acoustic radiated power (upper right), electric input power (lower left), and efficiency (lower right).

In Figure 5 and Figure 6, a series of characteristic loudspeaker driver curves are shown. In some figures the results obtained with the hybrid lumped and finite element of COMSOL are compared with theoretical curves that are based on the piston approximation.

In the upper left of Figure 5, the calculated speaker cone axial velocity u_D is shown together with two theoretical curves representing the approximate high-frequency and low-frequency behavior of a piston driver (see Ref. 1). The general trend is that the lowfrequency curve fits well for most of the frequency range. The high-frequency curve is seen to converge toward the higher frequencies. If you increase the frequency range of the model, the trend is that the high-frequency approximation is a better fit as expected.

The next two graphs in Figure 5 (top right and bottom left) represent the acoustically radiated power P_{AR} (see Equation 3) and the electric input power P_{E} (see Equation 4). At the low frequencies, the behavior is just as the piston model, as expected. The transition to nonmatching models occurs when the speaker diameter becomes comparable to half the wavelength at around 700 Hz. The last graph of Figure 5 represents the driver efficiency given in percent (%), that is, the ratio of the input electric power to the acoustic radiated

power. The actual efficiency is seen to match well with the predicted reference efficiency η_0 of about 1 %.

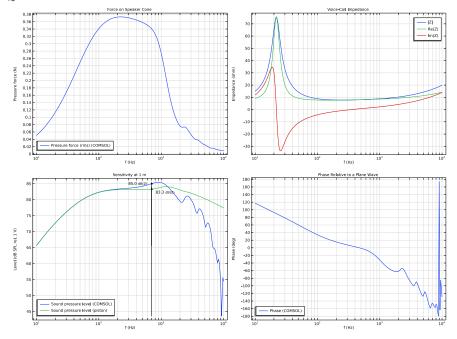


Figure 6: The frequency-dependent pressure force on the speaker cone (rms value) (upper left), voice coil impedance (upper right), speaker sensitivity (lower left), and phase response (lower right).

In Figure 6 (upper left), the RMS value of the acoustic pressure force F_D on the membrane (see Equation 1) is shown as function of frequency. The RMS value is obtained by the usual formula

$$\sqrt{0.5 \cdot F_{
m D} \cdot F_{
m D}^{\star}}$$

where the * represents the complex conjugate.

Figure 6 (top right) represents the voice coil impedance (absolute value, real, and imaginary part) calculated as $V_{\rm i}/i_c$. The resonance in the electric system is seen to coincide with the fundamental resonance frequency $F_{\rm S}$ = 22.0 Hz (see Table 3 and the Global **Definitions>Parameters** list in the model).

The two last figures in Figure 6 (bottom left and right) represent the speaker response (sound pressure level) measured 1 m in front and driven at 1 V RMS and the relative phase measured in the same point. The sensitivity is seen to match the piston model at the low frequencies as expected and it has realistic values for a speaker unit. The phase is represented as the phase of the pressure p(0,1 m) relative to the phase a plane wave would have in the same point $\exp(-ik_0z)$ with z = 1 m. In this way, the pure distance (phase lag) component of the phase has been removed.

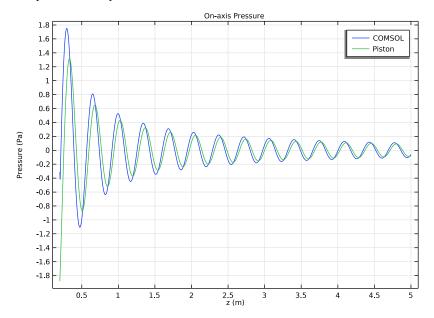


Figure 7: The pressure along the center z-axis from just outside the computational domain to a distance of $\bar{5}$ m. Evaluated for a frequency of 1000 Hz. The pressure is calculated using the exterior field calculation feature.

The pressure field along the z-axis is shown in Figure 7 from $z = R_{air}$ to z = 5 m, evaluated at a frequency of 1000 Hz. In the model, a Parameterized Curve 2D is used to evaluate the exterior-field pressure outside the computational mesh. The figure compares the modeled pressure with the analytical on axis pressure from a piston. The agreement is seen to be quite good at this frequency. For lower frequencies, the agreement becomes much better while the two curves diverge at higher frequencies, as expected (change the evaluation frequency in the plot to see this trend).

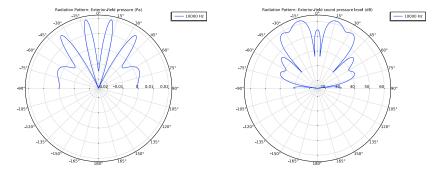


Figure 8: Exterior-field pressure and sound pressure level evaluated at a distance of 1 m the half sphere in front of the speaker and at 10 kHz.

The pressure and the sound pressure level evaluated at 1 m, using the dedicated radiation pattern plots, are shown in Figure 8, here evaluated at 10 kHz. In the figure, the 0° mark corresponds to the axial z direction. Both figures show a very strong directive pattern as expected at this high frequency.

In Figure 9 and Figure 10, the radiated intensity is illustrated and evaluated around the four frequencies 100 Hz, 1000 Hz, 5 kHz, and 10 kHz. The color plot represents the magnitude of the intensity vector **I**, the domain vector field represents the components of the intensity vector, and finally the vectors plotted on the edges represent the surface normals. The normals are useful when setting up, for example, the expression for the total radiated power (Equation 3). In this case, it is necessary to use the up() operator to get the intensity on the upper side of the speaker cone (seen relative to the normal direction). The four plots clearly show how the acoustic energy is focused for increasing frequencies.

The so-called directivity plot of the speaker unit is shown in Figure 11. This plot represents a contour plot of the sound pressure level $L_{
m p}$ evaluated along a half circle in front of the speaker as function of the angle and the frequency, that is, $L_p(\theta, f)$. The x-axis represents the angle and runs from -90° to 90°. The y-axis is a logarithmic frequency axis running from 10¹ Hz to 10⁴ Hz. The plot illustrates how the spatial response goes from a nearly omni-directional constant value at the low frequencies, through a single lobe response at intermediate frequencies, and ends up as a complex directive pattern at high frequencies. This type of plot is very often used, in industry, to characterize speakers and speaker units.

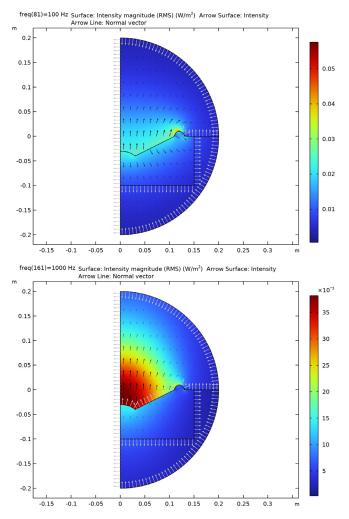


Figure 9: Intensity magnitude (color plot), intensity vector field (domain arrows), and surface normals (edge arrows) for $100~\rm{Hz}$ and $1000~\rm{Hz}$.

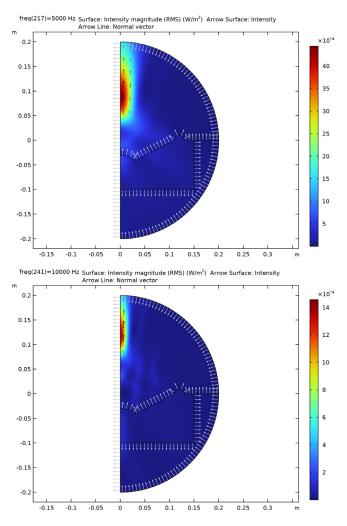


Figure 10: Intensity magnitude (color plot), intensity vector field (domain arrows), and surface normals (edge arrows) for $5~\rm kHz$ and for $10~\rm kHz$.

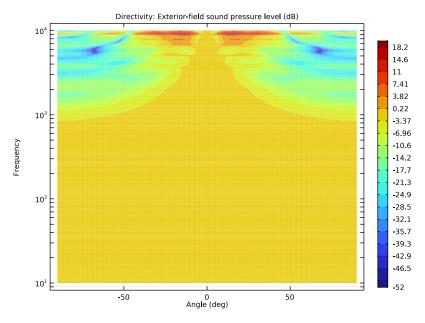


Figure 11: Directivity plot for the speaker. The x-axis is a scaled azimuthal angle that runs from -90° to 90° and the y axis is a logarithmic frequency axis that runs from 10^1 Hz to $10^4 Hz = 10 kHz$.

CLOSED BACK-VOLUME CONFIGURATION

When the model is solved with the closed back-volume configuration, the response of the system changes. This is easily seen in the voice coil impedance curves in Figure 12. The location of the fundamental resonance frequency F_s is seen to increase. This is because the presence of the closed space behind the speaker diaphragm corresponds to an increased mechanical stiffness. The sensitivity of the speaker is depicted in Figure 13. The presence of the closed volume is seen to heavily influence the low frequency performance of the speaker. This is typically also why speaker cabinets have a ventilation port, that is, to reduce this low frequency roll off.

Testing the model with a closed back-volume also has the advantage of testing the entire model setup (the electrical circuit to FEM coupling). For example, a sign error in the applied pressure force on the diaphragm would give more or less correct results for the open configuration, but would not predict the expected resonance shift. This kind of sanity tests are important when setting up numerical models.

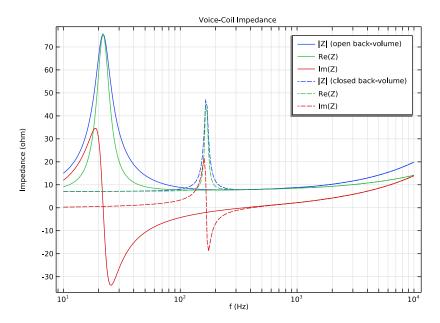


Figure 12: Voice coil impedance curves for both the open (solid) and closed (dashed) back-volume configurations.

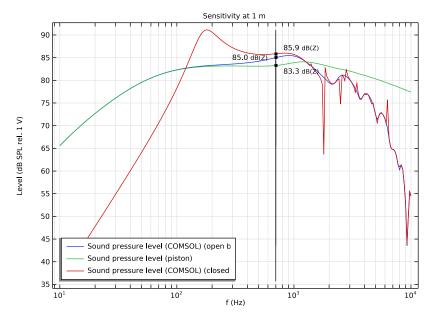


Figure 13: Sensitivity curves for the open back-volume configuration, the analytical results, and the closed back-volume configuration.

References

- 1. W. Marshall Leach, Jr., Introduction to Electroacoustics and Audio Amplifier Design, Kendall Hunt, 2010.
- 2. Loudspeaker Driver Model Documentation, from the COMSOL Application Library.

Application Library path: Acoustics_Module/Electroacoustic_Transducers/ lumped_loudspeaker_driver

Modeling Instructions

From the File menu, choose New.

NEW

In the New window, click Model Wizard.

MODEL WIZARD

- I In the Model Wizard window, click 2D Axisymmetric.
- 2 In the Select Physics tree, select Acoustics>Pressure Acoustics>Pressure Acoustics, Frequency Domain (acpr).
- 3 Click Add.
- 4 In the Select Physics tree, select AC/DC>Electrical Circuit (cir).
- 5 Click Add.
- 6 Click Study.
- 7 In the Select Study tree, select General Studies>Frequency Domain.
- 8 Click **Done**.

GLOBAL DEFINITIONS

Load all the model parameters from a file; they include all the small signal parameters (Table 2 and Table 3) as well as geometry parameters.

Parameters 1

- I In the Model Builder window, under Global Definitions click Parameters I.
- 2 In the Settings window for Parameters, locate the Parameters section.
- 3 Click Load from File.
- 4 Browse to the model's Application Libraries folder and double-click the file lumped_loudspeaker_driver_parameters.txt.

Build the simple 2D axisymmetric geometry of the speaker driver by drawing some circles and lines.

GEOMETRY I

Circle I (c1)

- I In the Geometry toolbar, click Circle.
- 2 In the Settings window for Circle, locate the Size and Shape section.
- 3 In the Radius text field, type 1[cm].
- 4 In the Sector angle text field, type 180.
- **5** Locate the **Position** section. In the **r** text field, type **a**.

Circle 2 (c2)

- I In the Geometry toolbar, click Circle.
- 2 In the Settings window for Circle, locate the Size and Shape section.
- 3 In the Radius text field, type Rair+Rpml.
- 4 In the Sector angle text field, type 180.
- **5** Locate the **Rotation Angle** section. In the **Rotation** text field, type -90.
- 6 Click to expand the Layers section. In the table, enter the following settings:

Layer name	Thickness (m)
Layer 1	Rpml

Polygon I (poll)

- I In the **Geometry** toolbar, click / Polygon.
- 2 In the Settings window for Polygon, locate the Object Type section.
- **3** From the **Type** list, choose **Open curve**.
- **4** Locate the **Coordinates** section. In the table, enter the following settings:

r (m)	z (m)
3[cm]	-4[cm]
a-1[cm]	0

Polygon 2 (pol2)

- I In the Geometry toolbar, click / Polygon.
- 2 In the Settings window for Polygon, locate the Object Type section.
- 3 From the Type list, choose Open curve.
- **4** Locate the **Coordinates** section. In the table, enter the following settings:

r (m)	z (m)
a+1[cm]	0
Rair	0

- 5 Click **Build All Objects**.
- **6** Click the **Zoom Extents** button in the **Graphics** toolbar.

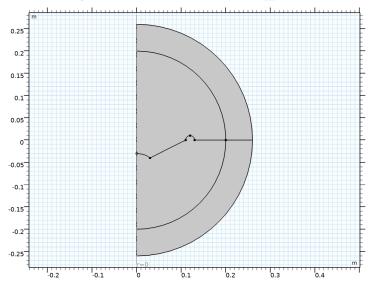
Quadratic Bézier I (qbI)

- I In the Geometry toolbar, click * More Primitives and choose Quadratic Bézier.
- 2 In the Settings window for Quadratic Bézier, locate the Control Points section.

- 3 In row 2, set r to 1.8[cm].
- 4 In row 3, set r to 3[cm].
- 5 In row 1, set z to -3[cm].
- 6 In row 2, set z to -3.1[cm].
- 7 In row 3, set z to -4[cm].
- 8 Locate the Weights section. In the 2 text field, type 1.5.
- 9 Click Build All Objects.

Delete Entities I (del I)

- I In the Model Builder window, right-click Geometry I and choose Delete Entities.
- 2 On the object c1, select Boundaries 3 and 4 only.
- 3 In the Settings window for Delete Entities, click **Build Selected.**



Polygon 3 (pol3)

- I In the Geometry toolbar, click / Polygon.
- 2 In the Settings window for Polygon, locate the Object Type section.
- 3 From the Type list, choose Open curve.

4 Locate the **Coordinates** section. In the table, enter the following settings:

r (m)	z (m)
0.15	0
0.15	-0.1
0	-0.1

5 Click Build All Objects.

Now, set up all the variables, selections, and component couplings under the **Definitions** node. Load the variables from the two variable files (one for model variables and one for the analytical piston expressions). The selections represent the speaker cone surface with an internal acceleration (red line in Figure 3) and the internal sound hard wall.

DEFINITIONS

Model variables

- I In the Home toolbar, click a= Variables and choose Local Variables.
- 2 In the Settings window for Variables, type Model variables in the Label text field.
- 3 Locate the Variables section. Click **Load from File**.
- **4** Browse to the model's Application Libraries folder and double-click the file lumped loudspeaker driver variables 1.txt.

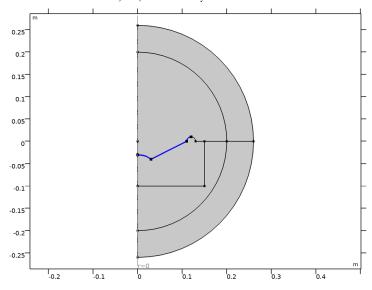
Analytical approximations

- I In the Home toolbar, click a= Variables and choose Local Variables.
- 2 In the Settings window for Variables, type Analytical approximations in the Label text field.
- 3 Locate the Variables section. Click **Load from File.**
- **4** Browse to the model's Application Libraries folder and double-click the file lumped_loudspeaker_driver_variables_2.txt.

Sbeaker

- I In the **Definitions** toolbar, click **\(\frac{1}{2} \) Explicit**.
- 2 In the Settings window for Explicit, type Speaker in the Label text field.
- 3 Locate the Input Entities section. From the Geometric entity level list, choose Boundary.

4 Select Boundaries 8, 15, and 18 only.



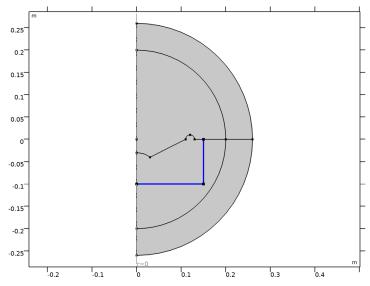
Baffle (interior wall)

- I In the **Definitions** toolbar, click **\(\bigcap_{\text{a}} \) Explicit**.
- 2 In the Settings window for Explicit, type Baffle (interior wall) in the Label text field.
- 3 Locate the Input Entities section. From the Geometric entity level list, choose Boundary.
- 4 Select Boundaries 9, 11, 12, and 19 only.

Back Volume (interior wall)

- I In the **Definitions** toolbar, click **\(\frac{1}{2} \) Explicit**.
- 2 In the Settings window for Explicit, type Back Volume (interior wall) in the Label text field.
- 3 Locate the Input Entities section. From the Geometric entity level list, choose Boundary.

4 Select Boundaries 4 and 10 only.



Perfectly Matched Layer I (pml1)

- I In the Definitions toolbar, click M. Perfectly Matched Layer.
- **2** Select Domains 1 and 5 only.
- 3 In the Settings window for Perfectly Matched Layer, locate the Scaling section.
- 4 From the Coordinate stretching type list, choose Rational.
- 5 In the PML scaling factor text field, type 0.5.
- 6 In the PML scaling curvature parameter text field, type 5.

You have now changed the default settings for the perfectly matched layer (PML). The new settings will improve the performance of the PML at very low frequencies. First of all, the acoustic radiated power should be positive. If the default settings had been used you would see a negative radiated power at low frequencies (see Figure 5 (top right) and the instructions on how to create the plot, further down). The issue is that at low frequencies the evanescent waves created by the moving speaker cone extend into the PML layer. The interaction between the scaled coordinate system in the PML and these waves may create an erroneous energy contribution to the model (can be either positive or negative). Note that the evanescent waves decay in only a fraction of a wavelength.

A good way to investigate the performance of the PML is to make a sensitivity analysis on some parameter (for example the total radiated acoustic power) with respect to

changes in the PML parameters. In this model we do not perform such a sensitivity analysis.

In general, increasing the curvature factor effectively shifts the resolving power of the PML toward the physical domain, which is necessary in this case since the evanescent components decay in only a fraction of a wavelength. However, if you increase it too much, you may lose resolution in the other end, that is, of the free space wavelength. Assuming that the PMLs work properly for high frequencies and curvature parameter 1, you can in principle do a convergence study increasing a constant curvature parameter until the low-frequency result converges (for 10 Hz) while making sure that the high-frequency result (for 10 kHz) is not affected. It turns out that a value of 5 yields good results in this model. A scaling factor of 0.5 further improves the results, but only by a small amount. Decreasing the scaling factor corresponds to compressing the PML layer (shortening it), which in turn effectively increases the mesh resolution.

ADD MATERIAL

- I In the Home toolbar, click **Add Material** to open the Add Material window.
- 2 Go to the Add Material window.
- 3 In the tree, select Built-in>Air.
- 4 Click Add to Component in the window toolbar.
- 5 In the Home toolbar, click **‡** Add Material to close the Add Material window. Now, set up the physics and the boundary conditions for the model. Use the interior conditions at the diaphragm and on the infinite baffle. This condition will allow for a discontinuous pressure field.

PRESSURE ACOUSTICS, FREQUENCY DOMAIN (ACPR)

Interior Sound Hard Boundary (Wall) I

- I In the Model Builder window, under Component I (compl) right-click Pressure Acoustics, Frequency Domain (acpr) and choose Interior Conditions> Interior Sound Hard Boundary (Wall).
- 2 In the Settings window for Interior Sound Hard Boundary (Wall), locate the **Boundary Selection** section.
- 3 From the Selection list, choose Baffle (interior wall).

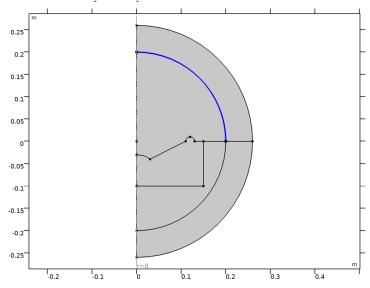
Interior Lumped Speaker Boundary 1

I In the **Physics** toolbar, click **Boundaries** and choose Interior Lumped Speaker Boundary.

- 2 In the Settings window for Interior Lumped Speaker Boundary, locate the Boundary Selection section.
- 3 From the Selection list, choose Speaker.

Exterior Field Calculation 1

- I In the Physics toolbar, click Boundaries and choose Exterior Field Calculation.
- **2** Select Boundary 16 only.



- 3 In the Settings window for Exterior Field Calculation, locate the Exterior Field Calculation section.
- 4 From the Condition in the $z = z_0$ plane list, choose Symmetric/ Infinite sound hard boundary.

Note that you have applied the exterior-field calculation condition only to the front of the speaker. In reality, the exterior-field condition should be applied to boundaries surrounding all sources and scatterers. This is of course not possible with an infinite baffle. However, in this specific case, a trick can be used, namely employing the fact that symmetry (in z = 0) is equal to a sound-hard wall in pressure acoustics (as indicated in the UI). So, in the infinite baffle configuration, the exterior-field condition can still be used.

Interior Sound Hard Boundary (Wall) 2

I In the **Physics** toolbar, click **Boundaries** and choose Interior Sound Hard Boundary (Wall).

- 2 In the Settings window for Interior Sound Hard Boundary (Wall), locate the Boundary Selection section.
- 3 From the Selection list, choose Back Volume (interior wall).

This interior wall condition will be deactivated in the first study when modeling the open configuration. In the second study the interior wall is active and the speaker operates with a closed back-volume.

Proceed to set up the electric circuit system for the electric and mechanical model. When building this look at Figure 3 for the references to the node numbers used in the model.

ELECTRICAL CIRCUIT (CIR)

In the Model Builder window, under Component I (compl) click Electrical Circuit (cir).

Voltage Source I (VI)

- I In the Electrical Circuit toolbar, click (2) Voltage Source.
- 2 In the Settings window for Voltage Source, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
Р	1
n	0

4 Locate the Device Parameters section. In the $v_{
m src}$ text field, type V0.

Resistor I (RI)

- I In the Electrical Circuit toolbar, click Resistor.
- 2 In the Settings window for Resistor, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
Р	1
n	2

4 Locate the **Device Parameters** section. In the R text field, type R g.

Resistor 2 (R2)

- I In the Electrical Circuit toolbar, click Resistor.
- 2 In the Settings window for Resistor, locate the Node Connections section.

3 In the table, enter the following settings:

Label	Node names
Р	2
n	3

4 Locate the **Device Parameters** section. In the R text field, type R E.

Inductor I (LI)

- I In the Electrical Circuit toolbar, click OOO Inductor.
- 2 In the Settings window for Inductor, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
Р	3
n	4

4 Locate the **Device Parameters** section. In the L text field, type L_E.

Resistor 3 (R3)

- I In the Electrical Circuit toolbar, click Resistor.
- 2 In the Settings window for Resistor, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
Р	3
n	4

4 Locate the **Device Parameters** section. In the R text field, type Rp E.

Current-Controlled Voltage Source I (HI)

- 2 In the Settings window for Current-Controlled Voltage Source, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
Р	4
n	0

4 Locate the **Device Parameters** section. In the **Gain** text field, type BL[m/Wb*ohm]. Remember to select the measured current when the mechanical circuit components have been set up.

Current-Controlled Voltage Source 2 (H2)

- 2 In the Settings window for Current-Controlled Voltage Source, locate the **Node Connections** section.
- **3** In the table, enter the following settings:

Label	Node names
Р	6
n	0

- **4** Locate the **Device Parameters** section. In the **Gain** text field, type BL[m/Wb*ohm].
- 5 Locate the Current Measurement section. From the Measure current for device list, choose Resistor 2 (R2).

Inductor 2 (L2)

- I In the Electrical Circuit toolbar, click QQQ Inductor.
- 2 In the Settings window for Inductor, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
P	6
n	7

4 Locate the **Device Parameters** section. In the L text field, type M MD[H/kg].

Current-Controlled Voltage Source I (HI)

- I In the Model Builder window, click Current-Controlled Voltage Source I (HI).
- 2 In the Settings window for Current-Controlled Voltage Source, locate the **Current Measurement** section.
- 3 From the Measure current for device list, choose Inductor 2 (L2).

Resistor 4 (R4)

- I In the Electrical Circuit toolbar, click Resistor.
- 2 In the Settings window for Resistor, locate the Node Connections section.

3 In the table, enter the following settings:

Label	Node names
Р	7
n	8

4 Locate the **Device Parameters** section. In the R text field, type R MS[ohm/kg*s].

Capacitor I (CI)

- I In the Electrical Circuit toolbar, click Capacitor.
- 2 In the Settings window for Capacitor, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
Р	8
n	9

4 Locate the **Device Parameters** section. In the C text field, type C MS[F*N/m].

External I vs. U I (IvsUI)

- I In the Electrical Circuit toolbar, click External I vs. U.
- 2 In the Settings window for External I vs. U, locate the Node Connections section.
- **3** In the table, enter the following settings:

Label	Node names
Р	9
n	0

4 Locate the **External Device** section. From the V list, choose Voltage from lumped speaker boundary (acpr/ilsb1).

The lumped electromechanical model defined by the electric circuit is now fully coupled with the Interior Lumped Speaker Boundary feature in the finite element domain.

MESH

Proceed and generate the mesh using the **Physics-controlled mesh** functionality. The frequency controlling the maximum element size is per default taken From study. Set the desired **Frequencies** in the study step. In general, 5 to 6 second-order elements per wavelength are needed to resolve the waves. For more details see Meshing (Resolving the

Waves) in the Acoustics Module User's Guide. In this model we use the default Automatic option, that gives 5 elements per wavelength.

MESH I

- I In the Model Builder window, under Component I (compl) click Mesh I.
- 2 In the Settings window for Mesh, locate the Physics-Controlled Mesh section.
- 3 In the table, clear the Use check box for Electrical Circuit (cir). Now, proceed to the study and set the frequencies, before building the mesh and solving.

STUDY I - OPEN BACK-VOLUME

- I In the Model Builder window, click Study I.
- 2 In the Settings window for Study, type Study 1 Open Back-Volume in the Label text field.

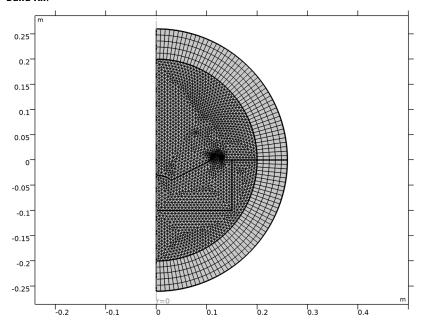
Step 1: Frequency Domain

- I In the Model Builder window, under Study I Open Back-Volume click Step 1: Frequency Domain.
- 2 In the Settings window for Frequency Domain, locate the Study Settings section.
- 3 Click Range.
- 4 In the Range dialog box, choose ISO preferred frequencies from the Entry method list.
- 5 In the Start frequency text field, type 10.
- 6 In the Stop frequency text field, type 10000.
- 7 From the Interval list, choose 1/24 octave.
- 8 Click Replace.

Disable the interior wall that defines the closed back-volume for this first study where we use the open configuration.

- 9 In the Settings window for Frequency Domain, locate the Physics and Variables Selection section.
- 10 Select the Modify model configuration for study step check box.
- II In the tree, select Component I (compl)>Pressure Acoustics, Frequency Domain (acpr)> Interior Sound Hard Boundary (Wall) 2.
- 12 Click / Disable.

MESH I In the Model Builder window, under Component I (compl) right-click Mesh I and choose **Build All.**



STUDY I - OPEN BACK-VOLUME

In the **Home** toolbar, click **Compute**.

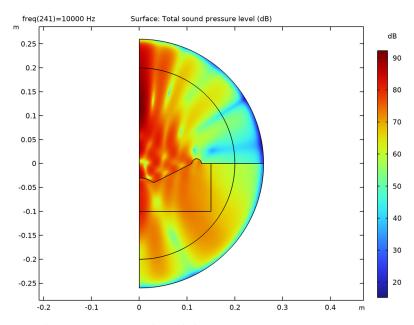
RESULTS

Acoustic Pressure (acpr)

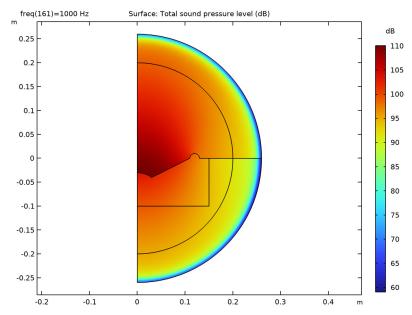
First, look at the default plots. Investigate the 2D Sound Pressure Level (acpr) plot to verify the performance of the perfectly matched layer (PML). After doing this you can disable plotting in the PML region, which is unphysical. Secondly, look at the default exterior-field plots and make a few changes, before setting up a range of plots to investigate the loudspeaker driver performance.

Sound Pressure Level (acpr)

Look at the sound pressure level (SPL) plots at the frequencies of 10 kHz, 1 kHz, and 10 Hz. Note that the SPL decreases nearly 100 dB over the width of the thickness of the PML. This means that the outgoing waves are extremely damped.

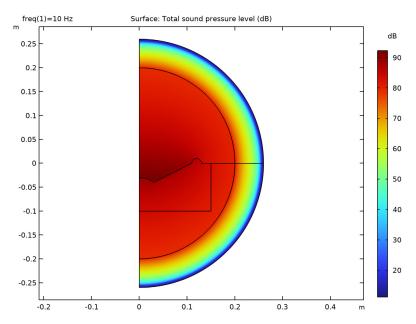


- I In the Model Builder window, click Sound Pressure Level (acpr).
- 2 In the Settings window for 2D Plot Group, locate the Data section.
- 3 From the Parameter value (freq (Hz)) list, choose 1000.



5 From the Parameter value (freq (Hz)) list, choose 10.

6 In the Sound Pressure Level (acpr) toolbar, click Plot.



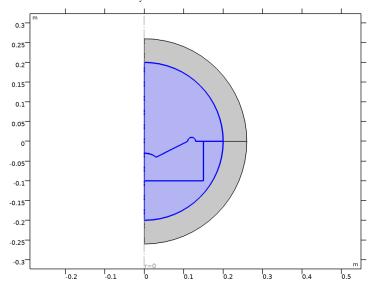
Study I - Open Back-Volume/Solution I (soll)

In the Model Builder window, expand the Results>Datasets node, then click Study I -Open Back-Volume/Solution I (soll).

Selection

- I In the Results toolbar, click has a Attributes and choose Selection.
- 2 In the Settings window for Selection, locate the Geometric Entity Selection section.
- 3 From the Geometric entity level list, choose Domain.

4 Select Domains 2–4 only.

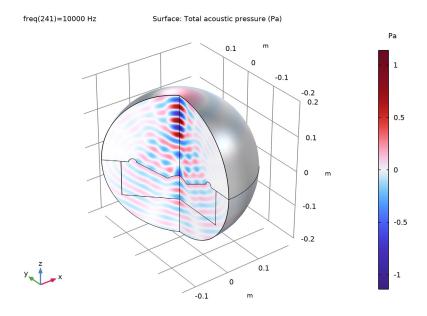


Acoustic Pressure (acpr)

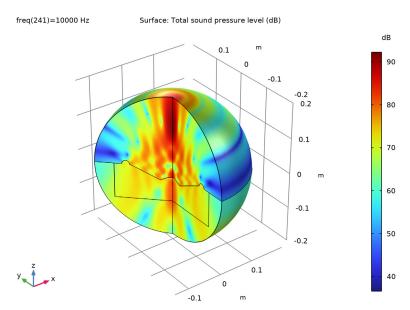
- I In the Model Builder window, under Results click Acoustic Pressure (acpr).
- 2 In the Settings window for 2D Plot Group, locate the Data section.
- 3 From the Parameter value (freq (Hz)) list, choose 1000.
- 5 Click the **Zoom Extents** button in the **Graphics** toolbar.
- 6 From the Parameter value (freq (Hz)) list, choose 5000.
- 7 In the Acoustic Pressure (acpr) toolbar, click **Plot**.

These two plots should reproduce Figure 4.

Acoustic Pressure, 3D (acpr)



Sound Pressure Level, 3D (acpr)



Radiation Pattern I

- I In the Model Builder window, expand the Exterior-Field Sound Pressure Level (acpr) node, then click Radiation Pattern I.
- 2 In the Settings window for Radiation Pattern, locate the Evaluation section.
- 3 Find the Angles subsection. From the Restriction list, choose Manual.
- 4 In the ϕ start text field, type -90.
- 5 In the ϕ range text field, type 180.

Radiation Pattern I

- I In the Model Builder window, expand the Exterior-Field Pressure (acpr) node, then click Radiation Pattern I.
- 2 In the Settings window for Radiation Pattern, locate the Evaluation section.
- 3 Find the Angles subsection. From the Restriction list, choose Manual.
- 4 In the ϕ start text field, type -90.
- 5 In the ϕ range text field, type 180.
- 6 In the Exterior-Field Pressure (acpr) toolbar, click **1** Plot.

These two polar plots should reproduce Figure 8.

Now, create nine 1D plots that depict various speaker characteristic plots. Each plot includes a number of steps to set up the title, the axis, the plot name and so on, these are optional and can be skipped. You can also just create the specific plot that is of interest to you. The next steps reproduce the graphs in Figure 5, Figure 6, and Figure 7.

The plots are the following:

- I Membrane Velocity
- **2** Acoustic Radiated Power
- **3** Electric Input Power
- **4** Efficiency
- **5** Force on Speaker Cone
- **6** Voice-Coil Impedance
- **7** Sensitivity
- 8 Phase
- 9 On-axis Exterior-Field Pressure

Diaphragm Velocity

- I In the Home toolbar, click Add Plot Group and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Diaphragm Velocity in the Label text field.
- 3 Click to expand the **Title** section. From the **Title type** list, choose **Label**.
- 4 Locate the Plot Settings section.
- **5** Select the **x-axis label** check box. In the associated text field, type f (Hz).
- 6 Select the y-axis label check box. In the associated text field, type Velocity amplitude (m/s).

Global I

- I Right-click Diaphragm Velocity and choose Global.
- 2 In the Settings window for Global, locate the y-Axis Data section.
- **3** In the table, enter the following settings:

Expression	Unit	Description		
abs(u_D)	m/s	COMSOL		
abs(Qlf_D/S_D)	m/s	Low frequency approximation		
abs(Qhf_D/S_D)	m/s	High frequency approximation		

- 4 In the Diaphragm Velocity toolbar, click **Plot**.
- 5 Click the x-Axis Log Scale button in the Graphics toolbar.

This plot should reproduce Figure 5 (top left).

Acoustic Radiated Power

- I In the Home toolbar, click Add Plot Group and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Acoustic Radiated Power in the Label text field.
- **3** Locate the **Title** section. From the **Title type** list, choose **Label**.
- 4 Locate the **Plot Settings** section.
- **5** Select the **x-axis label** check box. In the associated text field, type f (Hz).
- 6 Select the y-axis label check box. In the associated text field, type Power (W).
- 7 Locate the Legend section. From the Position list, choose Lower middle.

Global I

I Right-click Acoustic Radiated Power and choose Global.

- 2 In the Settings window for Global, click Add Expression in the upper-right corner of the y-Axis Data section. From the menu, choose Component I (compl)>Definitions> Variables>P_AR Radiated power (COMSOL) W.
- 3 Click Add Expression in the upper-right corner of the y-Axis Data section. From the menu, choose Component I (compl)>Definitions>Variables>P_AR_ana Radiated power (piston) W.
- 4 In the Acoustic Radiated Power toolbar, click Plot.
- 5 Click the x-Axis Log Scale button in the Graphics toolbar.
 This plot should reproduce Figure 5 (top right).

Electric Input Power

- I In the Home toolbar, click **Add Plot Group** and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Electric Input Power in the Label text field.
- 3 Locate the Title section. From the Title type list, choose Label.
- 4 Locate the **Plot Settings** section.
- **5** Select the **x-axis label** check box. In the associated text field, type **f** (Hz).
- 6 Select the y-axis label check box. In the associated text field, type Power (W).
- 7 Locate the Legend section. From the Position list, choose Lower right.

Global I

- I Right-click Electric Input Power and choose Global.
- 2 In the Settings window for Global, click Replace Expression in the upper-right corner of the y-Axis Data section. From the menu, choose Component I (compl)>Definitions> Variables>P_E Electric input power (rms) W.
- 3 In the Electric Input Power toolbar, click Plot.
- 4 Click the x-Axis Log Scale button in the Graphics toolbar.
 This plot should reproduce Figure 5 (bottom left).

Driver Efficiency

- I In the Home toolbar, click <a> Add Plot Group and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Driver Efficiency in the Label text field.
- 3 Locate the Title section. From the Title type list, choose Label.
- 4 Locate the Plot Settings section.
- **5** Select the **x-axis label** check box. In the associated text field, type **f** (Hz).

- 6 Select the y-axis label check box. In the associated text field, type Efficiency (%).
- 7 Locate the Legend section. From the Position list, choose Lower middle.

Global I

- I Right-click Driver Efficiency and choose Global.
- 2 In the Settings window for Global, locate the y-Axis Data section.
- **3** In the table, enter the following settings:

Expression	Unit	Description			
eta*100	1	Driver efficiency (COMSOL)			
eta0*100		Reference efficiency (piston)			

- 4 In the Driver Efficiency toolbar, click **Plot**.
- 5 Click the **x-Axis Log Scale** button in the **Graphics** toolbar.

This plot should reproduce Figure 5 (bottom right).

Force on Speaker Cone

- I In the Home toolbar, click **Add Plot Group** and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Force on Speaker Cone in the Label text field.
- 3 Locate the Title section. From the Title type list, choose Label.
- 4 Locate the Plot Settings section.
- **5** Select the **x-axis label** check box. In the associated text field, type **f** (Hz).
- 6 Select the y-axis label check box. In the associated text field, type Pressure force (N).
- 7 Locate the Legend section. From the Position list, choose Lower left.

Global I

- I Right-click Force on Speaker Cone and choose Global.
- 2 In the Settings window for Global, locate the y-Axis Data section.
- **3** In the table, enter the following settings:

Expression	Unit	Description
sqrt(0.5*F_D*conj(F_D))	N	Pressure force (rms) (COMSOL)

- 4 In the Force on Speaker Cone toolbar, click Plot.
- 5 Click the x-Axis Log Scale button in the Graphics toolbar.

This plot should reproduce Figure 6 (top left).

Voice-Coil Impedance

- I In the Home toolbar, click Add Plot Group and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Voice-Coil Impedance in the Label text field.
- 3 Locate the Title section. From the Title type list, choose Label.
- 4 Locate the Plot Settings section.
- **5** Select the **x-axis label** check box. In the associated text field, type f (Hz).
- 6 Select the y-axis label check box. In the associated text field, type Impedance (ohm).

Global I

- I Right-click Voice-Coil Impedance and choose Global.
- 2 In the Settings window for Global, locate the y-Axis Data section.
- **3** In the table, enter the following settings:

Expression	Unit	Description
abs(V0/cir.R1_i)	ohm	Z
real(VO/cir.R1_i)	ohm	Re(Z)
imag(VO/cir.R1_i)	ohm	Im(Z)

- 4 In the Voice-Coil Impedance toolbar, click Plot.
- 5 Click the x-Axis Log Scale button in the Graphics toolbar.

This plot should reproduce Figure 6 (top right).

Sensitivity at 1 m

- I In the Home toolbar, click In Add Plot Group and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Sensitivity at 1 m in the Label text field.
- 3 Locate the Title section. From the Title type list, choose Label.
- 4 Locate the Plot Settings section.
- 5 Select the x-axis label check box. In the associated text field, type f (Hz).
- 6 Select the y-axis label check box. In the associated text field, type Level (dB SPL rel. 1 V).
- 7 Locate the Legend section. From the Position list, choose Lower left.

Octave Band I

I In the Sensitivity at I m toolbar, click \to More Plots and choose Octave Band.

- 2 In the Settings window for Octave Band, locate the Selection section.
- 3 From the Geometric entity level list, choose Global.
- 4 Locate the y-Axis Data section. In the Expression text field, type pext(0,1).
- 5 Locate the Plot section. From the Quantity list, choose Continuous power spectral density.
- **6** Click to expand the **Legends** section. Select the **Show legends** check box.
- 7 From the Legends list, choose Manual.
- **8** In the table, enter the following settings:

Legends Sound pressure level (COMSOL)

Graph Marker I

- I Right-click Octave Band I and choose Graph Marker.
- 2 In the Settings window for Graph Marker, locate the Display section.
- 3 From the Display mode list, choose Line intersection.
- **4** In the **x-coordinates** text field, type 700.
- **5** Select the **Show lines** check box.
- 6 Locate the **Text Format** section. In the **Display precision** text field, type 3.
- 7 Select the **Include unit** check box.
- 8 Click to expand the Coloring and Style section. From the Anchor point list, choose Middle right.
- 9 In the Sensitivity at 1 m toolbar, click on Plot.

Octave Band 2

- I In the Model Builder window, under Results>Sensitivity at I m right-click Octave Band I and choose **Duplicate**.
- 2 In the Settings window for Octave Band, locate the y-Axis Data section.
- **3** In the **Expression** text field, type sqrt(2)*prms.
- **4** Locate the **Legends** section. In the table, enter the following settings:

Legends Sound pressure level (piston)

Graph Marker I

- I In the Model Builder window, expand the Octave Band 2 node, then click Graph Marker I.
- 2 In the Settings window for Graph Marker, locate the Coloring and Style section.

- 3 From the Anchor point list, choose Upper left.
- 4 In the Sensitivity at 1 m toolbar, click Plot.

This plot should reproduce Figure 6 (bottom left).

Phase Relative to a Plane Wave

- I In the Home toolbar, click . Add Plot Group and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Phase Relative to a Plane Wave in the Label text field.
- 3 Locate the Title section. From the Title type list, choose Label.
- 4 Locate the Plot Settings section.
- **5** Select the **x-axis label** check box. In the associated text field, type **f** (Hz).
- 6 Select the y-axis label check box. In the associated text field, type Phase (deg).
- 7 Locate the Legend section. From the Position list, choose Lower left.

Global I

- I Right-click Phase Relative to a Plane Wave and choose Global.
- 2 In the Settings window for Global, locate the y-Axis Data section.
- **3** In the table, enter the following settings:

Expression	Unit	Description
arg(subst(acpr.efc1.pext,r,0,z, 1[m])/(exp(-i*k0*1[m])*1[Pa]))	deg	Phase (COMSOL)

- 4 In the Phase Relative to a Plane Wave toolbar, click Plot.
- 5 Click the x-Axis Log Scale button in the Graphics toolbar.

This plot should reproduce Figure 6 (bottom right).

Set up a parameterized curve used to evaluate the exterior-field outside of the computational mesh.

Parameterized Curve 2D I

- I In the Results toolbar, click More Datasets and choose Parameterized Curve 2D.
- 2 In the Settings window for Parameterized Curve 2D, locate the Expressions section.
- 3 In the z text field, type s*5[m]+(1-s)*Rair.
- 4 Select the Only evaluate globally defined expressions check box.

On-axis Pressure

I In the Results toolbar, click \(\subseteq ID \) Plot Group.

- 2 In the Settings window for ID Plot Group, type On-axis Pressure in the Label text field.
- 3 Locate the Data section. From the Dataset list, choose Parameterized Curve 2D 1.
- 4 Locate the Title section. From the Title type list, choose Label.
- 5 Locate the **Plot Settings** section.
- 6 Select the x-axis label check box. In the associated text field, type z (m).
- 7 Select the y-axis label check box. In the associated text field, type Pressure (Pa).
- 8 Locate the Data section. From the Parameter selection (freq) list, choose From list.
- 9 In the Parameter values (freq (Hz)) list, select 1000.

Line Graph 1

- I Right-click On-axis Pressure and choose Line Graph.
- 2 In the Settings window for Line Graph, locate the y-Axis Data section.
- 3 In the Expression text field, type pext(r,z).
- 4 Locate the x-Axis Data section. From the Parameter list, choose Expression.
- 5 In the Expression text field, type z.
- 6 Click to expand the Quality section. From the Resolution list, choose Extra fine.
- 7 Click to expand the **Legends** section. Select the **Show legends** check box.
- 8 From the Legends list, choose Manual.
- **9** In the table, enter the following settings:

Legends COMSOL

Line Graph 2

- I Right-click Line Graph I and choose Duplicate.
- 2 In the Settings window for Line Graph, locate the y-Axis Data section.
- 3 In the Expression text field, type acpr.iomega*rho0*u D*S D*exp(-i*k0*z)/(2* pi*z).
- **4** Locate the **Legends** section. In the table, enter the following settings:

Legends **Piston**

5 In the On-axis Pressure toolbar, click Plot.

This plot should reproduce Figure 7.

Create a 2D intensity plot that includes the magnitude of the intensity vector acpr. I rms as well as an arrow surface (vector field plot) of the intensity vector, with the components (acpr.Ir,acpr.Iz).

Intensity

- I In the Home toolbar, click Add Plot Group and choose 2D Plot Group.
- 2 In the Settings window for 2D Plot Group, type Intensity in the Label text field.

Surface I

- I Right-click Intensity and choose Surface.
- 2 In the Settings window for Surface, locate the Expression section.
- 3 In the Expression text field, type acpr. I rms.

Arrow Surface 1

- I In the Model Builder window, right-click Intensity and choose Arrow Surface.
- 2 In the Settings window for Arrow Surface, click Replace Expression in the upper-right corner of the Expression section. From the menu, choose Component I (compl)> Pressure Acoustics, Frequency Domain>Intensity>acpr.lr,acpr.lz - Intensity.
- 3 Locate the Coloring and Style section. From the Color list, choose Black.

Arrow Line 1

- I Right-click Intensity and choose Arrow Line.
- 2 In the Settings window for Arrow Line, click Replace Expression in the upper-right corner of the Expression section. From the menu, choose Component I (compl)> Pressure Acoustics, Frequency Domain>Geometry>acpr.nr,acpr.nz - Normal vector.
- 3 Locate the Coloring and Style section. From the Color list, choose Gray.
- 4 In the Intensity toolbar, click Plot.
- **5** Click the **Zoom Extents** button in the **Graphics** toolbar.

This plot should reproduce the last frame in Figure 10, now change the evaluation frequency to 5000 Hz, 1000 Hz, and 100 Hz. This will reproduce the remaining frames in Figure 9 and Figure 10.

Intensity

- I In the Model Builder window, click Intensity.
- 2 In the Settings window for 2D Plot Group, locate the Data section.

- 3 From the Parameter value (freq (Hz)) list, choose 5000.
- 4 In the Intensity toolbar, click Plot.
- 5 From the Parameter value (freq (Hz)) list, choose 1000.
- 6 In the Intensity toolbar, click Plot.
- 7 From the Parameter value (freq (Hz)) list, choose 100.
- 8 In the Intensity toolbar, click **Plot**.

Next create the directivity plot of the speaker.

Directivity

- I In the Home toolbar, click Add Plot Group and choose ID Plot Group.
- 2 In the Settings window for ID Plot Group, type Directivity in the Label text field.

Directivity I

- I In the Directivity toolbar, click \to More Plots and choose Directivity.
- 2 In the Settings window for Directivity, locate the Evaluation section.
- 3 Find the Angles subsection. In the Number of angles text field, type 180.
- 4 From the Restriction list, choose Manual.
- 5 In the ϕ start text field, type -90.
- 6 In the ϕ range text field, type 180.
- 7 Click to expand the Coloring and Style section. From the Layout list, choose Frequency on y-axis.
- 8 In the Directivity toolbar, click Plot.
- **9** Click the **y-Axis Log Scale** button in the **Graphics** toolbar.

This should reproduce the directivity plot depicted in Figure 11. You can tailor the plot to your needs using the normalization options or defining the specific levels to use in the contour plot.

Create an additional plot that shows the sensitivity and the phase in one plot using two y-axes.

In the Model Builder window, under Results>Phase Relative to a Plane Wave right-click Global I and choose Copy.

Sensitivity and Phase

I In the Model Builder window, right-click Sensitivity at I m and choose Duplicate.

2 In the Settings window for ID Plot Group, type Sensitivity and Phase in the Label text field.

Octave Band 2

- I In the Model Builder window, expand the Sensitivity and Phase node.
- 2 Right-click Results>Sensitivity and Phase>Octave Band 2 and choose Delete.

Octave Band I

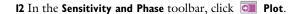
- I In the Model Builder window, under Results>Sensitivity and Phase click Octave Band I.
- 2 In the Settings window for Octave Band, locate the Plot section.
- 3 From the Quantity list, choose Band average power spectral density.
- 4 From the Band type list, choose 1/3 octave.

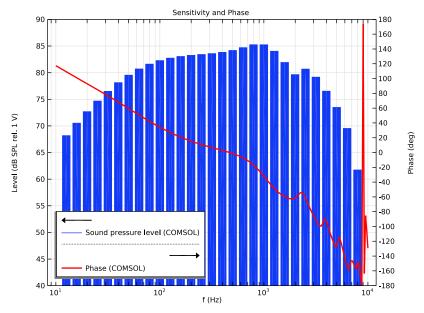
Global I

- I In the Model Builder window, right-click Sensitivity and Phase and choose Paste Global.
- 2 In the Settings window for Global, click to expand the Coloring and Style section.
- **3** From the **Width** list, choose **2**.
- 4 From the Color list, choose Red.

Sensitivity and Phase

- I In the Model Builder window, click Sensitivity and Phase.
- 2 In the Settings window for ID Plot Group, locate the Plot Settings section.
- 3 Select the **Two y-axes** check box.
- 4 In the table, select the Plot on secondary y-axis check box for Global 1.
- 5 Locate the Axis section. Select the Manual axis limits check box.
- **6** In the **y minimum** text field, type **40**.
- 7 In the y maximum text field, type 90.
- 8 In the Secondary y minimum text field, type -180.
- **9** In the **Secondary y maximum** text field, type 180.
- **10** Locate the **Plot Settings** section.
- II Select the Secondary y-axis label check box. In the associated text field, type Phase (deg).





It is time to solve the model with the closed back-volume configuration. After solving add a few extra graphs to the existing plots to see how the system response has changed. The closed back-volume acts as a spring and thus modifies the mechanical compliance of the system. The effect is, for example, seen in the electrical impedance plot by moving the resonance up in frequency.

ADD STUDY

- I In the Home toolbar, click Add Study to open the Add Study window.
- 2 Go to the Add Study window.
- 3 Find the Studies subsection. In the Select Study tree, select General Studies> Frequency Domain.
- 4 Click Add Study in the window toolbar.
- 5 In the Home toolbar, click Add Study to close the Add Study window.

STUDY 2

Step 1: Frequency Domain

- I In the Settings window for Frequency Domain, locate the Study Settings section.
- 2 Click Range.

- 3 In the Range dialog box, choose ISO preferred frequencies from the Entry method list.
- 4 In the Start frequency text field, type 10.
- 5 In the Stop frequency text field, type 10000.
- 6 From the Interval list, choose 1/24 octave.
- 7 Click Replace.
- 8 In the Model Builder window, click Study 2.
- 9 In the Settings window for Study, type Study 2 Closed Back-Volume in the Label text field.
- 10 Locate the Study Settings section. Clear the Generate default plots check box.
- II In the Home toolbar, click **Compute**.

Finally, modify two of the existing plots to understand the effect of a closed-back volume. To see the results of the closed configuration, in any of the plots, simply change the **Dataset** to the one from study 2.

RESULTS

Global I

- I In the Model Builder window, under Results>Voice-Coil Impedance click Global I.
- 2 In the Settings window for Global, locate the y-Axis Data section.
- **3** In the table, enter the following settings:

Expression	Unit	Description		
abs(V0/cir.R1_i)	ohm	Z (open back-volume)		

Global 2

- I Right-click Results>Voice-Coil Impedance>Global I and choose Duplicate.
- 2 In the Settings window for Global, locate the Data section.
- 3 From the Dataset list, choose Study 2 Closed Back-Volume/Solution 2 (sol2).
- 4 Locate the y-Axis Data section. In the table, enter the following settings:

Expression	Unit	Description		
abs(V0/cir.R1_i)	ohm	Z (closed back-volume)		

- 5 Locate the Coloring and Style section. From the Color list, choose Cycle (reset).
- **6** Find the **Line style** subsection. From the **Line** list, choose **Dashed**.

7 In the Voice-Coil Impedance toolbar, click Plot.

This plot should reproduce Figure 12.

Octave Band I

- I In the Model Builder window, under Results>Sensitivity at I m click Octave Band I.
- 2 In the Settings window for Octave Band, locate the Legends section.
- **3** In the table, enter the following settings:

Legends					
Sound pressure	level	(COMSOL)	(open	back-volume)	

Octave Band 3

- I Right-click Results>Sensitivity at I m>Octave Band I and choose Duplicate.
- 2 In the Settings window for Octave Band, locate the Data section.
- 3 From the Dataset list, choose Study 2 Closed Back-Volume/Solution 2 (sol2).
- **4** Locate the **Legends** section. In the table, enter the following settings:

Legends				
Sound pressure 1	Level	(COMSOL)	(closed	back-volume)

Graph Marker I

- I In the Model Builder window, expand the Octave Band 3 node, then click Graph Marker I.
- 2 In the Settings window for Graph Marker, locate the Coloring and Style section.
- 3 From the Anchor point list, choose Lower left.
- 4 In the Sensitivity at 1 m toolbar, click Plot.

This plot should reproduce Figure 13.