Measurement of complex impedance of ultrasonic transducers

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Abstract

The automated system for ultrasonic transducer complex impedance measurement using the sine wave correlation for signal amplitude and phase extraction is presented. The system contains the direct digital synthesis (DDS) generator and two ADC channels with a common reference frequency source. In such a case the frequency error between the excitation and the measurement channels is reduced. The shielded measurement chamber with local low noise high impedance preamplifiers has been added to reduce the induced noise and to improve robustness. The measurement procedure and equations are presented. The results of various ultrasonic transducers impedance measurements are given.

Keywords: Ultrasonic transducer impedance, data acquisition, complex impedance measurement, sine fitting, sine correlation.

Introduction

The transducer electrical impedance affects an ultrasonic transducer noise performance [1], driving response [2], bandwidth and sensitivity [3]. Having the ultrasonic transducer impedance response over the frequency band, the mentioned influence can be estimated. The complex impedance allows for even better judgment. Therefore there is a need for ultrasonic transducer complex impedance measurement. The techniques for measurement of a transducer impedance employ an external generator and the analog-to-digit converter (ADC) for voltage and current measurements [3-5]. The technique presented is using the sine wave correlation for signal amplitude and phase extraction. The aim of investigation presented is to inspect the suitability of the designed system for ultrasonic transducer complex impedance frequency response measurements.

Measurement setup

The use of a separate generator and ADC will impose the frequency error which will have to be compensated, using the advanced fitting techniques in order to determine the frequency [6,7]. If both the generator reference frequency and the ADC reference frequency use the same source, the mentioned problems are significantly reduced. We have decided to use the complete AC parameters measurement system [8]. The system contains the direct digital synthesis (DDS) generator and two ADC channels Ch1 and Ch2 with a common reference oscillator. The measurement system systematic error sources we are the following [5]:

- the reference impedance $R_{ref}$;
- the ADC board input impedances;
- the acquisition channel nonlinearity;
- the parasitic impedances in grounding, shielding and cabling.

Therefore the shielded measurement chamber with local low noise high impedance preamplifiers has been added to the system to reduce the induced noise and to improve the robustness of measurements. The connections diagram and the structure of the internal measurement chamber is presented in Fig.1.

![Fig. 1. Measurement setup](image1)

The measured impedance $Z_x$ can be calculated as:

$$Z_x = \frac{U_x}{I_x},$$

(1)

where $U_x$ is the complex voltage on measured impedance taps, $I_x$ is the complex current flowing in measured
impedance. Using the notation used in Fig.2, the $U_x$ and $I_x$ are given by:

$$U_x = \frac{U_{ADC1}}{R_{ref}},$$

$$I_x = \frac{U_{ADC2} - U_{ADC1}}{R_{ref}},$$

The measurement procedure is very simple and consists of the following steps:

(i) the exciting generator is adjusted to produce a sine wave with the desired measurement frequency and amplitude;

(ii) the acquisition channels simultaneously acquire data records corresponding to the voltages amplitude; 

(iii) voltages $U_{ADC1}$ and $U_{ADC2}$ in measurement channels introduced due to gain mismatch are measured at shorted $R_{ref}$ and removed the $Z_x$ condition.

(iv) voltages $U_{ADC1}$ and $U_{ADC2}$ due to gain mismatch are calculated and corrected for;

(v) the measured impedance $Z_x$ is calculated:

$$Z_x = \frac{U_{ADC1}}{U_{ADC2} - U_{ADC1}} \cdot R_{ref}.$$

The main random error source is the noise present in any system. The use of sine-fitting techniques can largely reduce the influence of a noise in final results [5]. Since we use the same reference frequency for DDS used to generate the exciting signal and the acquisition ADC, the more simple form of sine-fitting can be used. Then the wave to be fit is defined as:

$$u(t) = A \cos(2\pi ft) + B \sin(2\pi ft) + C,$$

where $A$ and $B$ are the in-quadrature sine amplitudes of the wave, $C$ is the DC component and $f$ is the excitation frequency used. Fitting this function to the set of $M$ samples, $y_1...y_M$, acquired at a frequency $f_s$ at time instances $t_1...t_M$, is accomplished by seeking the minimum of approximation error root-mean-square (RMS) value:

$$e_{RMS} = \frac{1}{M} \sum_{m=1}^{M} [y_m - (A \cos(2\pi f t_m) + B \sin(2\pi f t_m) + C)]^2.$$

The procedure is simple and iterative, thus consuming a lot of a computational time. The other approach could be to use a non-iterative procedure. This requires only the construction of a $M \times 3$ matrix and determination of its pseudo-inverse matrix and multiplication with the actual samples. The least-squares estimated parameters are obtained from

$$x = [D^T D]^{-1} D^T y = D^* y,$$

where $x$ is the estimated parameter vector, $D$ is the matrix that linearly relates the estimated parameters and the samples $y$. $D^*$ is the Moore–Penrose pseudoinverse matrix.

Then

$$x = \begin{bmatrix} A \\ B \\ C \end{bmatrix},$$

$$D = \begin{bmatrix} w(f, t_1) & g(f, t_1) \\ w(f, t_2) & g(f, t_2) \\ 5 & 5 \\ w(f, t_M) & g(f, t_M) \end{bmatrix},$$

The closer look at the operations performed indicate the close similarity of the operations performed to a correlation procedure, inherent in the Fourier transform. Therefore the procedure can be presented in a simplified way which is more computational efficient and is closer to programming conventions in a signal processing:

$$A = \frac{\sum_{m=1}^{M} \cos(2\pi f t_m) \cdot y_m}{\sum_{m=1}^{M} \cos^2(2\pi f t_m)},$$

$$B = \frac{\sum_{m=1}^{M} \sin(2\pi f t_m) \cdot y_m}{\sum_{m=1}^{M} \sin^2(2\pi f t_m)},$$

$$C = \frac{\sum_{m=1}^{M} y_m}{M}.$$

Then the measured signal magnitude is

$$U = \sqrt{A^2 + B^2},$$

and the phase

$$\phi = \arctan\left(\frac{B}{A}\right).$$

The same computational procedure has to be performed on both ADC channels Ch1 and Ch2.

The systematic and random errors, constituting accuracy estimation of the AC parameters measurement system [8] have been investigated in [10]. The investigation included theoretical equations analysis, simulation and real system experiments. The obtained estimations of a system precision can be used in evaluating the results obtained using the aforementioned acquisition system.

The measurement results

The AC parameters measurement system [8] has been used to obtain several ultrasonic transducers impedances.

Fig. 3. Air coupled 200 kHz transducer impedance $|Z_d|$ (1) and the phase $\varphi_x$ (2) vs. frequency
Air-coupled, 200 kHz operating frequency ultrasonic transducer impedance has been measured. The impedance magnitude $|Z|_x$ and the phase $\phi_x$ response are presented in Fig.3. The real and imaginary parts of the transducer impedance have been used to calculate the magnitude and the phase frequency response. Radial and secondary resonances are visible in the obtained impedance magnitude and phase plots in Fig.3.

It would be interesting to use the measured complex impedance to obtain the parameters of the ultrasonic transducer model. In [1] ultrasonic system noise performance has been evaluated using the Butterworth-Van Dyke (BVD) transducer model. Therefore the BVD model with 1 and 2 serial resonant tanks have been obtained, using complex impedance response fitting to the measured $Z_x$. The Matlab procedure \texttt{fmins}, employing the Nelder-Mead simplex method [11] have been used for fitting. Three $Z_{\text{approx}}$ fitting convergence rules have been used:

(i) Approximation error RMS value of the real and imaginary impedance parts:

$$
\varepsilon_{\text{Re RMS}} = \sqrt{\frac{1}{N} \sum_{n=1}^{N} \left| \text{Re}(Z_{x,n}) - \text{Re}(Z_{\text{approx},n}) \right|^2}, \quad (10a)
$$

$$
\varepsilon_{\text{Im RMS}} = \sqrt{\frac{1}{N} \sum_{n=1}^{N} \left| \text{Im}(Z_{x,n}) - \text{Im}(Z_{\text{approx},n}) \right|^2}, \quad (10b)
$$

$$
\varepsilon_{\text{RMS}} = \varepsilon_{\text{Re RMS}} + \varepsilon_{\text{Im RMS}}; \quad (10c)
$$

(ii) Approximation error RMS value of the impedance magnitude:

$$
\varepsilon_{\text{Re RMS}} = \sqrt{\frac{1}{N} \sum_{n=1}^{N} \left| Z_{x,n} - Z_{\text{approx},n} \right|^2}, \quad (11)
$$

(iii) Approximation error RMS value of the impedance phase:

$$
\varepsilon_{\text{RMS}} = \sqrt{\frac{1}{N} \sum_{n=1}^{N} \left| \phi_{Z_{x,n}} - \phi_{Z_{\text{approx},n}} \right|^2}. \quad (12)
$$

Here $N$ is the total number of impedance frequency response points, $n$ is the point number.

The real and imaginary $Z_x$ parts (solid lines) of the same air-coupled 200kHz transducer and their approximations using BVD model with 1 and 2 serial resonant tanks (dashed and dotted lines) are plotted in Fig.4.

The obtained BVD model parameters are presented in Table 1. Approximation with one serial resonant tank and using the Eq. 10 for a convergence rule is plotted by dashed lines (noted as 1STe10 in Table 1). Approximation with one serial resonant tank and using the Eq. 11 for convergence rule is noted as 1STe11 in Table 1. The one serial resonant tank and the Eq. 12 as application result of a convergence rule is plotted by dotted lines (notation 1STe12 in Table 1). Use of 2 tanks for transducer modeling and Eq. 12 as convergence rule is plotted by the dash-dot lines (2STe12 in Table 1). The results of application of Eq. 10 and Eq. 11 are very similar, as can be seen by analysis of the obtained BVD parameters in Table 1. Therefore, of Eq. 11 result of application for a convergence rule is not shown in Fig.4.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>1STe10</th>
<th>1STe11</th>
<th>1STe12</th>
<th>2STe12</th>
</tr>
</thead>
<tbody>
<tr>
<td>$C_\text{a}$, pF</td>
<td>138.10</td>
<td>137.80</td>
<td>77.094</td>
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<tr>
<td>$R_{S1}$, Ω</td>
<td>2053.7</td>
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<td>$C_{S1}$, pF</td>
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<td>40.976</td>
<td>21.291</td>
<td>28.569</td>
</tr>
<tr>
<td>$L_{S1}$, mH</td>
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<td>15.8</td>
<td>29.8</td>
<td>22.2</td>
</tr>
<tr>
<td>$R_{S2}$, Ω</td>
<td>4673.3</td>
<td></td>
<td></td>
<td></td>
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<td>$C_{S2}$, pF</td>
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<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>$L_{S2}$, mH</td>
<td>20.5</td>
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</table>

The presented BVD model fitting to the measured $Z_x$ results is only the demonstration of how the measurement results can be employed further. Other models can be successfully fit into measurement results, yielding better precision. Also new mechanical or acoustical parameters can be evaluated based on a model fitting.

The impedance measurement procedure is automated and results can be obtained quickly. As a demonstration of possibilities, a PZT ceramic disk impedance frequency response has been measured while changing one surface damping conditions. The disk impedance frequency response is presented in a complex coordinate system in Fig.5 and as real and imaginary impedance parts in Fig.6.

![Fig. 5. Transducer $Z_x$ polar plot in a complex coordinate system](image)

The impedance frequency response of the unloaded disk was measured first (solid line). Then one disk face...
was loaded with a light foam damper and the impedance measured (the dashed line). The impedance measurement results obtained by loading one disk face with a high viscosity latex damper are plotted by the dotted curve. The same results are presented as real and imaginary impedance parts in Fig. 6.

Fig. 6. PZT disk impedance variation with damping

Such investigation would prove useful during the design or manufacturing process of an ultrasonic transducer. The impedance can be verified after every processing stage.

The radio frequency (RF) transformers are extensively used in ultrasonic equipment [12]. By modification of source impedance the amplifier noise optimal impedance can be matched. This improves the noise performance of an ultrasonic system. The transformer also allows the effective amplifier input DC biasing thanks to a winding inductance.

If matching transformer is not properly chosen, degradation of the frequency response performance might take place. Therefore application of the matching transformer should be verified by impedance or AC response measurements. The impedance measurement results presented in Fig. 7 indicate the case when a matching transformer is used.

Fig. 7 Impedance modification by matching transformer

The phoenix SSW 70° 2MHz transducer impedance has been measured (the dashed curve). Then a matching transformer with turns ratio 2 has been added. The higher turns section has been connected to the transducer and the lower turns have been connected to the measurement system. The measurement results indicate there was no frequency response performance degradation due to implementation transformer. Note the same phase response in the matched and unmatched case.

Conclusions

The sine wave correlation for extraction of signal amplitude and phase is used. The use of the same frequency reference source for the exciting generator and the signals acquisition allows for significant simplification of a measurement procedure. The goal of the investigation presented is to indicate the multitude of applications of the obtained ultrasonic transducer complex impedance.

References

13. L. Svilainis,V. Dumbrava

Ultragarsinių keitiklių kompleksinio impedancio matavimas

Rezultatai