Time domain attenuation estimation method from ultrasonic backscattered signals

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Ultrasonic attenuation is important not only as a parameter for characterizing tissue but also for compensating other parameters that are used to classify tissues. Several techniques have been explored for estimating ultrasonic attenuation from backscattered signals. In the present study, a technique is developed to estimate the local ultrasonic attenuation coefficient by analyzing the time domain backscattered signal. The proposed method incorporates an objective function that combines the diffraction pattern of the source/receiver with the attenuation slope in an integral equation. The technique was assessed through simulations and validated through experiments with a tissue mimicking phantom and fresh rabbit liver samples. The attenuation values estimated using the proposed technique were compared with the attenuation estimated using insertion loss measurements. For a data block size of 15 pulse lengths axially and 15 beamwidths laterally, the mean attenuation estimates from the tissue mimicking phantoms were within 10% of the estimates using insertion loss measurements. With a data block size of 20 pulse lengths axially and 20 beamwidths laterally, the error in the attenuation values estimated from the liver samples were within 10% of the attenuation values estimated from the insertion loss measurements.

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I. INTRODUCTION

Ultrasonic wave propagation in random media at wavelengths comparable to or greater than the size of typical heterogeneities have been widely studied, often with a view toward characterization of the underlying microstructure. Quantitative ultrasound (QUS) techniques have been used to estimate the size, shape and mechanical properties of tissue microstructure for specified regions of interest (ROIs). Measurements of speed of sound, backscatter energy, scatterer size, and scatterer strength can provide a means for material characterization. However, many of these parameters require compensation for losses due to frequency-dependent attenuation. In and of itself, ultrasonic attenuation can also be used to distinguish between diseased and normal tissues. To this end, researchers have compiled attenuation coefficients for various tissues in different animals.

Several methods have been developed to estimate local attenuation in scattering media from ultrasonic backscatter. Generally these attenuation algorithms can be divided into three categories: (1) frequency domain, (2) time-frequency domain, and (3) time domain techniques. In the frequency domain techniques, various methods such as spectral shift, spectral difference, spectral-log difference, and hybrid method have been developed and assessed. Insana et al. modified the spectral difference method by accounting for transducer beam diffraction pattern using an empirically determined correction factor. It accounted for transducer size, focus, and band-pass characteristics. The hybrid method estimates attenuation coefficients by using the spectral difference technique and Gaussian filters to measure the downshift of the central frequency. The hybrid method is an improvement over the spectral difference technique by employing a Gaussian filter to estimate center frequency of the backscattered power spectra. The hybrid method was demonstrated to be more stable at boundaries with variations in backscatter. The hybrid method uses the advantages of the spectral difference and spectral shift techniques while minimizing their disadvantages. Labayed and Bigelow recently compared three spectral domain attenuation estimation methods, i.e., spectral difference, spectral log difference, and hybrid methods, using backscattered signals from simulated phantoms with varying scatterer sizes and concentrations. The authors investigated the error associated with using different data block sizes for accurate attenuation estimation.

Frequency domain techniques use the backscattered power spectrum to estimate attenuation coefficients. The backscattered power spectrum can be modeled as a multiplication of a system response function, a scattering function and an attenuation function. By using a reference signal the system dependent function can be compensated in the power spectrum, yielding the normalized power spectrum. However, the normalized power spectrum is still a combination of the scattering function and attenuation, which are both frequency dependent. By assuming that the scattering function is constant over a particular depth, the attenuation can be estimated by comparing normalized...
power spectra at different depths. Therefore, the axial resolution of this technique can suffer because power spectra taken from several depths are required to accurately estimate the attenuation. All the spectral domain techniques assume that the diffraction effects can be decoupled from attenuation effects, which may result in biased estimates when the effects due to diffraction and attenuation are comparable to each other.

To better deal with diffraction effects, researchers developed a reference phantom spectral domain technique that compared the backscatter signal from a sample to a known reference sample, i.e., a phantom, to estimate backscatter and attenuation of the sample.\textsuperscript{16,22} The reference technique accounts for system-dependent effects with the assumption that the speed of sound in the reference and the unknown sample are similar, otherwise diffraction effects will not be accounted for accurately. Researchers have developed algorithms to estimate backscatter coefficient and attenuation simultaneously.\textsuperscript{23,24} Bigelow et al.\textsuperscript{25} estimated backscatter and attenuation simultaneously using a Gaussian transformation and spectral fit algorithm. Nam et al.\textsuperscript{24} developed a technique to simultaneously estimate BSC and attenuation slopes using a least squares method. The authors used a power law model for BSCs and developed a three-parameter model to incorporate backscatter and attenuation coefficients. Then a least squares technique was employed to estimate backscatter and attenuation coefficients simultaneously. The authors used a glass bead phantom to compare theoretical and experimental results. The authors modeled the frequency dependence of the backscatter coefficients using a power law fit to estimate both the parameters simultaneously.

Researchers have also investigated estimating attenuation based on time-frequency domain techniques. Fink et al.\textsuperscript{25} used a short time Fourier transform to calculate the down shift of the spectral centroid with depth to estimate attenuation. Again diffraction effects had been decoupled from attenuation in the estimation process.

In one time domain approach, He and Greenleaf used the monotonic relationship between the signal-to-noise ratio of the echo envelope peaks and the variance of the mean power of the scattered echoes to estimate attenuation coefficients.\textsuperscript{26,27} Jang et al.\textsuperscript{28} estimated attenuation based on the entropy difference between two adjacent envelope segments of narrowband ultrasound pulse echoes. The authors used segments of size 10 mm\,×\,10 mm from a depth of 3 cm using a 2.5 MHz transducer. Knipp and co-workers estimated ultrasound attenuation from B-mode image data.\textsuperscript{29} The authors used a “gray-scale look-up table” to convert image pixel value data within an ROI to echo signal amplitudes and compare these to echo signals from the same depth in a reference phantom. These relative echo levels were used to estimate attenuation in the respective ROI. Rice introduced the relationship between the density of zero crossings of a waveform and the power spectrum of that waveform.\textsuperscript{30,31} Flax et al.\textsuperscript{32} demonstrated that the density of zero crossings could be related to attenuation coefficients by assuming the waveform could be described by a Gaussian spectrum. Again all the time domain methods discussed in the preceding text estimated attenuation by decoupling diffraction effects from attenuation effects. The time domain backscattered signal is a convolution of the source/receiver characteristics, scattering function, and attenuation function. Therefore the assumption of separating the source/receiver characteristics from the convolution integral may break down when diffraction effects are comparable with attenuation effects in the backscattered signal.

Time domain approaches can be easier to implement and can be computationally more cost effective compared to frequency or time-frequency domain techniques. However, it is more difficult to account for diffraction correction in time domain techniques. Recently, Ghoshal et al.\textsuperscript{33,34} modeled the backscatter response in the time domain using a stochastic wave equation where the material properties and source/receiver characteristics were coupled in an integral equation. The authors modeled the diffraction pattern using a single-order Gaussian beam model and compared the theoretical model with experimental data using high f-number transducers to extract grain size in polycrystalline media. The authors suggested that a single-order Gaussian beam model might not model the beam pattern accurately away from the focus specifically for transducers with high f-numbers. Thus a better model for the transducer beam pattern was necessary to successfully account for the backscatter response for a wide variety of transducer geometries.

In this study, the theoretical model developed by Ghoshal et al.\textsuperscript{33,34} was modified using O’Neil’s\textsuperscript{35} analytical solution for the axial pressure for a focused transducer to estimate attenuation coefficients from tissues. Using an iterative process, the attenuation coefficient could be estimated by solving an integral equation where the attenuation coefficient and the diffraction of the beam pattern were coupled. A second model was derived by approximating the transducer axial pressure by multiple Gaussian functions, and a closed form solution for the time domain backscattered response was obtained. The attenuation coefficient was estimated for different data blocks corresponding to different locations imaged from a sample. The theory and the objective function for the optimization problem are briefly discussed in Sec. II. The experimental methods and results from tissue-mimicking phantoms and fresh rabbit liver samples are provided in Sec. III. Finally, Sec. IV provides some conclusions regarding the study.

II. THEORY

Consider a block of data that corresponds to several consecutive lines of range gated ultrasonic backscattered signal from a sample. The variance of all the signals from a single data block is given by

\[
\Phi(t) = \langle V(t)^2 \rangle - \langle V(t) \rangle^2
\]

where the angular bracket \(\langle y \rangle\) is defined as the mean of \(y\), and \(V(t)\) is the backscattered signal corresponding to the data block. Using a single scattering assumption and a single order Gaussian beam pattern, the variance of signals in a given data block of axial length \(z_1\) to \(z_2\) is given by\textsuperscript{33,34}
where \( w_0 \) is the transducer element radius, \( w(z) \) is the beam width as function of depth \( z \) in the respective material. The distance between the transducer surface and the sample, i.e., the water path, is denoted by \( z_p \), the distance \( z_i \) is from the start of the surface of the solid media to the start of the data block, and the axial distance \( z = z_f + (c/c_f)z_i \). If no water path exists between the sample and the source, then \( z_p = 0 \) and \( z_i = z \). The local attenuation coefficient, mass density, and speed of sound in the sample are denoted by \( \sigma \), \( \rho \), and \( c \), respectively. Similarly, the attenuation coefficient, mass density and speed of sound in water are denoted by \( \rho_w \), \( \rho_f \), and \( c_f \), respectively. The total attenuation from the sample surface to the start of the data block is denoted by \( \eta \). The time domain input wave is assumed to be a Gaussian pulse where \( \sigma \) is the width in the time domain. It is assumed the backscatter coefficient denoted by \( \sigma_b \) is spatially and temporally independent and denoted at the center frequency associated with the input wave \( \omega_0 \). In practice, to extract the system dependent response, a reference signal can be acquired by reflecting a pulse from a planar reflector placed at the focus of the transducer. The peak amplitude of the reflected reference signal from the planar reflector placed at the focus of transducer (distance from the transducer face and the focus is denoted by \( z_f \)) is denoted by \( V_{\text{max}} \). The diffraction constant \( D(0,0,2\text{REF}) \) is used to account for the diffraction effects of the reference signal. The reflection coefficient of the planar reflector is denoted by \( R_p \), and the transmission coefficient from the water to the material and from sample to water are denoted by \( T_{bl} \) and \( T_{lf} \), respectively.

The displacement field using a complex single-order Gaussian beam can be written as

\[
\psi(x, t) = A_0 \exp(ikz) \times \frac{q(0)}{q(z)} \exp\left[-ik_f \frac{x^2 + y^2}{2q(z)}\right],
\]

where \( k = \omega_0/c \) is the wave number and \( A_0 \) is the amplitude at \( x = y = z = 0 \). The complex Gaussian beam parameter is defined by

\[
1 - \frac{2}{R(z)} = \frac{1}{q(z)} \left[ \frac{1}{R(z)} + i \frac{2k_f}{w^2(z)} \right],
\]

where \( q(z) \) is a function of beamwidth \( w(z) \) and radius of curvature \( R(z) \) along axial direction of the propagating beam. The Gaussian beamwidth and the radius of curvature of the wavefront may be written as

\[
w^2(z) = \frac{2 \sigma_t}{k_f \text{Im}(1/q(z))}, \quad R(z) = \frac{1}{\text{Re}(1/q(z))}.
\]

At \( z = 0 \), the beamwidth is the transducer radius \( w(0) = w_0 \), the radius of curvature of the wavefront is \( R(0) = -F \), and \( F \) is the focal length. The source field can be separated into its amplitude \( A(x) \) and phase \( \Theta(x) \) as

\[
\psi(x) = A(x) \exp(i\Theta(x)),
\]

where the amplitude is given by

\[
A(x) = A_0 \frac{w_0}{w(z)} \exp\left[-\frac{x^2 + y^2}{w^2(z)}\right],
\]

and the phase terms can be deduced accordingly. From Eqs. (4) and (5), the factor \( w_0^2/w^2(z) \) may be written as

\[
\frac{w_0^2}{w^2(z)} = \frac{1 + \beta^2}{\left( \frac{z}{F} \right)^2 + \left( 1 + \beta^2 \right)^2}. \tag{6}
\]

where \( \beta = 2F/(k_f w_0^3) \) and the wave vector is defined by \( k_f = \omega_0/c_f \). From Eq. (3), the factor \( w_0^2/w^2(z) \) is the square of the normalized axial pressure amplitude of the transducer for a single order Gaussian beam model. Next, the axial pressure of a focused transducer beam is approximated using two different models.

### A. O’Neil model

O’Neil’s model has provided the analytical solution for the axial pressure for a focused transducer as long as the focusing is not too severe using a continuous wave as the forcing function. Therefore, from O’Neil, Eq. (8) can be written as

\[
\frac{w_0^2}{w^2(z)} = \left| \frac{m(z)}{m(z)} \right| \frac{\left| \exp(jk_f z) - \exp(jk_f r_e) \right|^2}{4 m^2(z) \sin^2(0.5k_f(z - r_e))}, \tag{9}
\]

where \( |y| \) denotes absolute of \( y, m(z) = 1 - z/F \), \( r_e = \sqrt{(z - h)^2 + a^2} \), and \( h = \sqrt{F^2 - a^2} \). Field II was used to simulate a transducer field pattern in water for the sets of transducers shown in Table I. The comparison of the O’Neil model and Field II results are shown in Figs. 1(a)–1(c). The pressure field along the transducer axis was normalized by the maximum strength of the field and compared with the O’Neil model from Eq. (9) as shown in Fig. 1(b). It can be observed that the difference between the theory and the simulated axial pressure is negligible.

Combining Eqs. (2) and (9), the single scattered response is given by

\[
\Phi(t) = \phi_0 \int_{z_1}^{z_2} \frac{4 \sin^2(0.5k_f(z_i - r_e(z_i)))}{m^2(z_i)} \times \exp\left(-4\pi z - \frac{(2z - ct)^2}{\sigma^2c^2}\right) dz, \tag{10}
\]

where \( \phi_0 \) is a constant obtained from Eq. (2). Generally data blocks are constructed around the focal region of the...
Using the multiple Gaussian function the single scattered response is given by

\[ \Phi(t) = \phi_0 \sum_{p=1}^{M} A_p \exp \left[ -\frac{(z_p - z_f)^2}{\gamma_p^2} \right] \times \exp \left[ -4xz - \frac{(2z - c_t)^2}{\sigma^2c^2} \right] dz. \]  

Here the limits of integration were changed compared to the limits from Eq. (10), such that the analytical solution can be obtained. This change of limits does not violate any assumption because the Gaussian time domain pulse causes the integrand to approach zero depending upon \( t \) and the input wave pulse length, i.e., the location of data block. Therefore this model may be used to estimate local attenuation for different depths in the sample by adjusting the analysis region in terms of time \( t \). Using the multiple Gaussian function to model the transducer axial pressure, the time domain backscattered response is given by

\[ \Phi(t) = \phi_0 \sum_{p=1}^{M} A_p \exp \left[ -\frac{\gamma_p^2c^2\sigma^2}{C_0^2c^2+c_0^2} \right] \times \exp \left[ -\frac{4(z_p - z_f)^2 + 4\alpha(z_p - z_f)c_r^2\sigma^2 - 4c_r^2\sigma^2\gamma_p^2}{4\gamma_p^2c_r^2\sigma^2} \right] \times \exp \left[ -\frac{-c_r^2\sigma^2 - 2\gamma_p^2}{4\gamma_p^2c_r^2\sigma^2} \right]. \]  

where \( c_r = c/c_f \) is the ratio of speed of sound in the sample and water (assumed to couple the transducer to the sample). The preceding equation is an important result of the current study where a closed formal solution is derived for time domain backscattered response. In this result, the attenuation function and the transducer characteristics are coupled. The attenuation coefficient is estimated by minimizing the mean squared error between the model [from Eq. (13)] and experimental results \( \Phi_{\text{EXP}}(t) \). Mathematically the optimization problem is stated as

\[ \alpha = \arg \min_x \frac{1}{Q} \sum_{i=1}^{Q} (\Phi(t_i, x) - \Phi_{\text{EXP}}(t_i))^2, \]  

where \( \Phi_{\text{EXP}}(t) \) is the variance of all signals in the respective data block obtained from backscattered time-domain signals.
\[ Q \] is the total number of time samples used in the data block and \( U_N(t_i; a) \) is the theoretical normalized backscatter response at time \( t_i \) using attenuation coefficient of \( a \). The normalization is obtained using

\[
U_N(t) = \frac{U(t)}{\phi_0 \sum_{i=1}^{Q} \Phi_{\text{EXP}}(t_i)},
\]

(15)

Therefore to estimate attenuation, \( \phi_0 \) is not required to be calculated. The preceding normalization procedure adjusts the magnitude of the theoretical model to be comparable to the magnitude of the experimental result. The attenuation value used for \( \Phi(t_i; a) \) that minimizes the least mean square error between the model and the experimental data is considered the attenuation value for the respective data block. The attenuation coefficient is converted from Np/cm to dB/cm.

All the results presented in this study were obtained by using the multiple Gaussian functions and closed form solution of the time domain backscatter response. For all the transducer geometries used for this study, four Gaussian functions were used to model the axial pressure profile of each transducer.

III. EXPERIMENTAL METHODS AND RESULTS

A. Phantom construction

Two tissue-mimicking phantoms (Phantoms A and B) used by Wear et al.\(^{40}\) with different BSCs and attenuation coefficients were used here. Glass beads were used as scatterers in all the phantoms embedded in agarose, n-propanol, condensed milk, and water. Phantoms A and B had glass bead size distributions of 75–90 and 9–43 \( \mu \)m, respectively. The construction process of the Phantoms A and B was described by Wear et al.\(^{40}\)

Simulations were conducted to test the new technique. The software phantoms were constructed in MATLAB (The Mathworks Inc., Natick, MA) and signals generated using Field II.\(^{38,39}\) Three-dimensional volume matrices were constructed of 50 mm in the direction parallel to the transducer face, 2 mm in height, and of various depths (65 and 80 mm). The source consisted of a sinusoidal pulse with center frequency of 5 MHz and a pulse length of approximately 1.5 cycles. The beamwidth (\(-6\) dB) was 900 \( \mu \)m for the source, assuming a Gaussian beamwidth for an \( f/3 \) aperture. The beam was focused at different depths in the phantom by changing the aperture size of the source/receiver and keeping the \( f/ \)number of 3 as a constant. Point scatterers were placed at random spatial locations in the phantom volume using a uniform probability density function.

B. Liver samples

Fresh liver samples were extracted from male New Zealand white rabbits acquired from Myrtle’s Rabbitry (Tompson’s Station, TN). The rabbits had been on a special fatty diet.\(^{41}\) The basal diet contained 10% fat, 1% cholesterol, 0.11% Mg, 14% protein, and 54% carbohydrates (Catalog No. 1811279, 5TZB, Purina Test Diet, Richmond, IN). Therefore the rabbit livers had a higher fat content than a normal rabbit liver. Ultrasonic experiments were conducted within 15 min of the removal of the liver from the body.

C. Ultrasonic methods

A schematic of the experimental setup is shown in Fig. 3. The sample was completely submerged and a single-element focused transducer was used for scanning. The specifications for the focused transducers used for the experiments are listed in Table 1. For scanning, the tissue-mimicking phantoms were immersed in a tank of degassed water and the liver samples were immersed in 0.9% saline solution made from degassed water. The transducers were operated in pulse-echo mode using a Panametrics 5900 pulser/receiver (GE Panametrics, Inc., Waltham, MA), and the echo signals were recorded and digitized with a 14 bit, 200 MHz A/D card (Model: PDA14-200, Signatec, Newport Beach, CA) and downloaded to a PC computer for postprocessing. The sampling rate used to digitize the received signals was 200 MHz. In an experiment, the sample was held stationary and the transducer was moved using a computer-controlled micropositioning system (Daedal, Inc., Harrisburg, PA). For analysis, rf signals backscattered from the samples were recorded. A step size of approximately one-half beamwidth was used between consecutive scan lines. The temperature of the water bath was maintained at room temperature for the phantom experiments. A mechanical coil heater (Waage Electric Inc., Kenilworth, NJ)
controlled by a water bath temperature controller (Model YS172, Yellow Spring Instrument Co., Inc., Yellow Spring, OH) was used to maintain water bath temperature at 37 °C for the experiments with liver samples.

Sound speed and attenuation were also estimated in the phantoms and liver samples using time of flight and insertion loss methods, respectively. To estimate sound speed, arrival times of received pulses were measured with and without the sample in the water path between the transducer and a planar Plexiglas reflector. The sound speed in the sample was computed from

\[
c = c_f \left( \frac{t_3 - t_1}{t_2 - t_1} \right),
\]

where \( c_f \) is the speed of sound in the surrounding fluid media, \( t_3 \) is the propagation time to the reflector measured when no sample is in the path, and \( t_1 \) and \( t_2 \) are the arrival times of the frontwall of the sample and the sample/reflector interface echoes, respectively. The propagation times were estimated using cross-correlation.

Using the speed of sound estimated from Eq. (16), the sample thickness \( D \) was calculated. The attenuation coefficient was estimated using

\[
\alpha(f) = -\frac{1}{2D} \ln \left( \frac{|F_2(f)|}{|F_3(f)|} \right),
\]

where \( |F_2(f)| \) and \( |F_3(f)| \) are the frequency response of the time domain reflected signal from the Plexiglas reflector with and without the sample in the path, respectively. The attenuation coefficients estimated from the insertion loss methods at the center frequency of the input wave were compared with the results obtained from the proposed time domain attenuation estimation approach.

D. Experimental results

The variance of all signals from a sample of thickness \( d \) is shown in Fig. 4, where all the peaks are defined explicitly. The transducer pulse-echo signal was collected at several different spatial positions by translating the transducer across the sample. From Fig. 4, the peak between the frontwall and the backwall was mainly due to the material properties and source/receiver characteristics. The theoretical model shown in Eq. (13) can be used to fit the experimental data to estimate various parameters.

First, the input signal was analyzed in a region with uniform single scattering and fit to a Gaussian envelope to estimate \( \sigma \). An example signal reflected from a planar surface placed at the focus of transducer \( f/4-8 \) MHz and the corresponding best-fit Gaussian envelope is shown in Fig. 5(a), and its frequency response is shown Fig. 5(b). Here the pulse length (PL) is referred to as the duration of the pulse [shown in Fig. 5(a)], where the amplitude falls from its max value by 20 dB.

Next, the variances of the signal envelopes were calculated from a data block. The signal envelopes were used instead of the raw signals because this helped in reducing the high frequency fluctuations. Example variances of the signals for two different data block sizes are shown in Figs. 6(a) and 6(b). As expected, more averaging of scan lines reduced the high frequency fluctuations in the variance curve. An example B-mode of Phantom B and its parametric map of attenuation are shown in Figs. 6(c) and 6(d), respectively. The attenuation map was constructed using a data block size of 20 PL axially and 20 beam widths (BW) laterally. For all the results presented here, the data blocks were overlapped by 90% in both axial and lateral directions for the purposes of creating parameter images.

From insertion loss measurements the estimated attenuation coefficients for Phantom A and B were \( 1.85 \pm 0.025 \) and \( 4 \pm 0.35 \) dB/cm, respectively, at 5 MHz. Wear et al. published insertion loss measurement from the same Phantom A and B as \( 0.4 \) and \( 0.7 \) dB/cm/MHz, respectively, in the frequency range of 2.5–5 MHz, which corresponds to 2 and

![Image](image-url)
FIG. 6. (Color online) (a) Example variance of the signal and the model fit for a single data block of axial and lateral lengths of 20 PL and 20 BW, respectively. (b) Example variance of the signal and the model fit for a single data block of axial and lateral lengths of 25 PL and 50 BW respectively. (c) B-mode image, and (d) parametric B-mode image enhanced by attenuation maps for Phantom B.

FIG. 7. (Color online) Attenuation estimated for Phantom A in terms of (a) absolute percent error and (b) standard deviation error and for Phantom B in terms of (c) absolute percent error (d) standard deviation error for the estimates for various combinations of data block sizes using transducer f/3–5 MHz.
3.5 dB/cm for Phantom A and B, respectively, at 5 MHz. The interlaboratory comparison of the attenuation coefficients from Phantom A and B resulted in agreement within 0.75 and 1 dB/cm, respectively at 5 MHz compared to the insertion loss measurements, between the participating laboratories for the study.43

The absolute percent errors in mean attenuation coefficients were estimated using

\[
\text{ABS ERR} (\%) = 100 \times \frac{x_{\text{EXP}} - x_{\text{IL}}}{x_{\text{IL}}},
\]

where \(x_{\text{EXP}}\) and \(x_{\text{IL}}\) refers to the estimated mean attenuation using the new method and insertion loss measurements. Error curves for phantoms A and B using transducer \(f/3\text{--}5\) MHz are shown in Figs. 7(a) and 7(c), respectively. The percentage standard deviation error was estimated using

\[
\text{STD} (\%) = 100 \times \frac{x_{\text{STD}}}{x_{\text{MEAN}}},
\]

where \(x_{\text{MEAN}}\) and \(x_{\text{STD}}\) are the mean and standard deviation of the estimated attenuation from all the data blocks. The percentage standard deviation error from Phantoms A and B using transducer \(f/3\text{--}5\) MHz are shown in Figs. 7(b) and 7(d), respectively. The bias and the standard deviations decreased as the axial length of the data block size increased. The estimated percentage error and percentage standard deviation error using transducer \(f/4\text{--}8\) MHz are shown in Figs. 8(a)--8(d). Similarly, the results using transducer \(f/2\text{--}7\) MHz from Phantoms A and B are shown in Figs. 9(a)--9(d). Due to a small depth of field, the maximum possible axial data block size was 15 PL. Overall as the block size was increased both in the lateral and axial direction the error in estimated attenuation decreased compared to the insertion loss measurements. The new method yielded attenuation estimates within 10% of insertion loss measurements at a data block size of 15 PL axially and 15 BW laterally.

To compare the results obtained using the proposed method with other methods, attenuation coefficients for Phantoms A and B were estimated from peaks of echo envelope as proposed by He and Greenleaf.26 The algorithm described in Sec. III of Ref. 11 was used, where the central 25 A-lines of the B-scan and the first 45 peaks were detected from the signal envelopes, which approximately corresponded to a data block size of 30--35 PL axially by 13 BW laterally. The comparison of the results from envelope peak method, and the new methods using two different data block size are tabulated in Table II. The new method resulted in similar estimates in terms of error and standard deviations for half the axial length of the data block size used in the envelope peak method. At similar data block sizes, the new method resulted in much lower bias and variance estimates compared to the envelope peak method.

Using simulated phantoms, attenuation coefficients were estimated at various depth from weakly and strongly
attenuating media. The attenuation estimates from three different numerical phantoms with attenuation coefficients of 2.5, 5, and 7.5 dB/cm at 5 MHz are shown in Fig. 10, using a data block size of 20 PL axially and 20 BW laterally. The attenuation coefficients estimated using the new method and without correcting for the downshift in the center frequency resulted in errors less than 10% compared to the actual attenuation values. Therefore the method appears to be robust at various depths in an attenuating media. However, it should be noted that for the results at 7.5 dB/cm (at 5 MHz) the signal-to-noise ratio was not sufficient to estimate attenuation accurately beyond 48 mm using the proposed method.

Attenuation coefficients were estimated from the rabbit liver samples using the new technique and compared with the estimates obtained by insertion loss using transducer f/3–11 MHz. From the insertion loss measurements, the estimated attenuation for the liver samples RL1 and RL2 were 7.81 ± 1.01 and 7.35 ± 1.12 dB/cm, respectively, at 11 MHz. Attenuation coefficients were estimated at different depths using the new technique for a data block size of 20 PL axially and 20 BW laterally. At each depth of the sample, the mean and standard deviation of the attenuation values were estimated from all the data blocks at that particular depth. The mean and standard deviation of attenuation values estimated using the new method and the insertion loss measurements are shown in Figs. 11 and 12 for liver samples RL1 and RL2, respectively. The mean attenuation values of 8.04 ± 1.03 and 7.19 ± 0.79 dB/cm/MHz were estimated for the selected data blocks from samples RL1 and RL2.

### TABLE II. Comparison of estimated error and standard deviations of attenuation coefficients from two different methods using transducer f/3–5 MHz.

<table>
<thead>
<tr>
<th></th>
<th>Phantom A</th>
<th>Phantom B</th>
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<tbody>
<tr>
<td></td>
<td>ABS ERR (%)</td>
<td>STD (%)</td>
</tr>
<tr>
<td>Envelope peak (31PL × 13BW)</td>
<td>10.81</td>
<td>24.18</td>
</tr>
<tr>
<td>New method (15PL × 15BW)</td>
<td>9.82</td>
<td>20.86</td>
</tr>
<tr>
<td>New method (30PL × 15BW)</td>
<td>0.25</td>
<td>9.64</td>
</tr>
</tbody>
</table>

FIG. 10. Attenuation using the new method estimated from simulated phantoms with mean attenuation coefficients of 2.5, 5, and 7.5 dB/cm at 5 MHz denoted by ○, □ and *, respectively, along the axial direction from the surface of the phantom.
respectively. The B-mode image and a parametric image enhanced by attenuation coefficients for the rabbit liver sample RL1 are shown in Figs. 13(a) and 13(b), respectively.

IV. CONCLUSION

A new time-domain attenuation estimation algorithm was introduced. The new technique coupled the spatial diffraction pattern of the source/receiver beam characteristics with the attenuation of the sample using a stochastic wave equation. The theoretical model derived here clearly demonstrated that it is the axial pressure of the beam that couples with the attenuation of the material. The axial pressure was modeled using O’Neil’s analytical solution and using multiple Gaussian functions. Using the Gaussian model, a closed form analytical solution for the time domain backscattered response was derived. Attenuation coefficients were then estimated using a numerical optimization scheme and multiple Gaussian functions to approximate the axial profile of the focused transducer beam from backscatter data from regions approximated by single and uniform scattering. Coupling the source/receiver characteristics with the attenuation function represents a significant improvement over previous time-domain attenuation techniques. The closed form solution obtained using multiple Gaussian functions was computationally more efficient compared to the solution obtained using O’Neil’s model because it did not require solving an integral equation numerically.

The attenuation values estimated using the new technique from tissue-mimicking phantoms for data block sizes of 15 PL axially by 15 BW laterally, agreed to within 10% of insertion loss estimates. Similarly the estimated attenuation from fresh rabbit liver samples, for data block sizes of 20 PL axially by 20 BW laterally, agreed to within 10% of insertion loss estimates. The results obtained for the fresh rabbit liver samples agreed to within one standard deviation of the insertion loss measurements.

For a tightly focused transducer ($f/2$), the new method also resulted in less than 10% error for data block sizes of 15 PL axially and 15 BW laterally. The error in the estimates decreased with increasing data block size but did not provide better than 5% error for the largest data block size. Interestingly, a larger data block size was required for the liver samples for less than 10% error with respect to the insertion loss estimates. The attenuation coefficients estimated using the new technique were compared with another time-domain technique based on envelope detection. The attenuation estimated using the new technique resulted in using similar effective data block sizes compared to some of the frequency domain techniques such as spectral-log difference and hybrid methods. In other frequency domain techniques, such as hybrid method, the backscatter coefficient is modeled as a power of frequency; this restricts its applicability. This contrasts with the new method proposed here that eliminates such limitations. Simulation and experimental results demonstrated that at a moderate depth of penetration and low to high attenuating media, the assumption resulted in errors of less than 10% of the expected values.

Frequency domain techniques, such as the spectral centroid shift method, estimate attenuation by comparing the downshift in the center frequency from several data blocks along the axial direction. The new technique resulted in using similar effective data block sizes compared to some of the frequency domain techniques such as spectral-log difference and hybrid methods. In other frequency domain techniques, such as hybrid method, the backscatter coefficient is modeled as a power of frequency; this restricts its applicability. This contrasts with the new method proposed here that eliminates such limitations. Simulation and experimental results demonstrated that at a moderate depth of penetration and low to high attenuating media, the assumption resulted in errors of less than 10% of the expected values.

Due to the assumption of the maximum energy content at the center frequency, the backscatter coefficient just modifies the amplitude of the variance of the backscattered signals rather than its shape. To neglect this assumption, the
backscatter coefficient can be integrated over the frequency band of the transducer but still would act as a constant amplitude term in the time domain variance function. However, the advantage of applying the technique without adjusting for the downshift in center frequency is that attenuation coefficients were obtained for each data block without knowing the backscatter coefficient.

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