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Fox, Paul D.; Cheng, Jiqi; Lu, Jian-yu

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Fourier-Bessel Field Calculation and Tuning of a CW Annular Array

Paul D. Fox, Jiqi Cheng, and Jian-yu Lu, *Senior Member, IEEE*

Abstract-A 1-D Fourier-Bessel series method for **computing and tuning the linear lossless field of flat continuous wave (CW) annular arrays is given and discussed with both numerical simulation and experimental verification. The technique provides a new method for modelling and manipulating the propagated field by linking the quantized surface pressure profile to a set** of **limited diffraction Bessel beams propagating into the medium. In the limit, these become a known set of nondiffracting Bessel beams satisfying the lossless linear wave equation, which allow us to derive a linear matrix formulation for the field in terms of the ring pressures on the transducer surface. Tuning (beamforming)** of **the field then follows by formulating a least squares design with respect to the transducer ring pressures. Results are presented in the context of a 10-ring annular array operating at 2.5 MHz in water.**

I. INTRODUCTION

 Γ ^N THIS PAPER, we describe a method for computing and Ituning linear lossless fields from flat CW annular arrays using **1-D** Fourier-Bessel series **[l], [Z].** The use of these series allows the propagated field to be described as a set of J_0 Bessel beams $[3]$, $[4]$, giving a linear mapping between the spatial ring pressures on the transducer surface and the propagated field at any point in space. Bessel beams have already been extensively studied **[5]-** [8], and this work builds on previous knowledge to draw up a method for both computing and tuning (beamforming) the propagated field by using a set of Bessel beam basis functions. In **[9],** these were applied across the transducer surface to decompose the emitted field into a known set of limited diffraction Bessel beams. In this paper, we extend the analysis to include the entire plane beyond the outer edge of the transducer, which, in the limit, describes the emitted field as a set of known nondiffracting Bessel beams. This feature allows us to apply analytic techniques to solve for the field **as** a weighted set of exact Bessel solutions to the wave equation, constituting a new field analysis tool complimenting other approaches such as **[lo]-[14].** We show that the method correlates well with both previous experimental results **(51** and simulations based 011

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P. *Fox* is with 0rsted,DTU, Technical University of Denmark, DK-2800 Lyngby, Denmark (e-mail: pdf@oersted.dtu.dk).

J. Cbeng and *LY.* Lu **are** with the Ultrasound Laboratory, Department of Bioengineering, University of Toledo, Toledo, OH 43606-3390 USA (e-mail: jilu@eng.utoledo.edu).

the (slower) Rayleigh-Sornmerfeld diffraction formula. The method also allows us to tune the field in **a** least squares sense with respect to **a** given desired field distribution. This has been introduced briefly through sirnulatioms in [15], modelling the tuning of a Bessel beam on an equal area annular array. In this paper, we extend the method to different tuning designs and an alternative array geometry **[51.**

Section I1 gives model definitions for the governing wave equation and structure of the CW annular arrays considered. Section **111** introduces the application and interpretation of 1-D Fourier-Bessel series in the context of annular arrays. In Section IV, we explain how to compute the propagated field using Fourier-Bessel series, with a numerical example provided in Section V. Sectiom VI then extends the analysis of the field propagation to derive a least squares beamforming design, with numerical examples given in Section VII. Finally, in Section VIII, we summarize, draw conclusions, and suggest further work.

11. MODEL DEFINITIONS

A. Propagation Model

Annular arrays have circular symmetry around the propagation axis, for which we assume the governing circular symnietric linear wave equation

$$
\left[\frac{1}{r}\frac{\partial}{\partial r}\left(r\frac{\partial}{\partial r}\right) + \frac{\partial^2}{\partial z^2} - \frac{1}{c^2}\frac{\partial^2}{\partial t^2}\right]f(r, z, t) = 0 \qquad (1)
$$

in which $f(r, z, t)$ is the field pressure relative to static pressure, *T* is the radial distance from the cylindrical centerline, *t* is the outward propagation distance perpendicular to the transducer surface (placed at $z = 0$ and centered around $r = 0$, and c is the speed of sound (assumed real). For a given wavenumber $k = \omega/c$, this equation has Bessel beam solutions [4], [7] of the form

$$
f(r, z, t) = J_0(\alpha r) \cdot e^{j\beta z} \cdot e^{-j\omega t}
$$

\n
$$
\alpha^2 + \beta^2 = k^2.
$$
\n(2)

B. Annular **Arrays**

We then consider N-ring flat annular arrays with monochromatic surface pressure $q(r, t) = q(r)e^{-j\omega t}$, in which $q(r)$ is the spatial quantization profile and ω is the angular frequency. In annular arrays, $q(r)$ is stepwise constant with discrete quantization values q_p , where

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 $p = 1, \ldots, N$ is the ring number, and $p = 1$ for the inner ring with $p = N$ for the outer ring. Inner and outer ring radii are denoted r_p^- and r_p^+ , respectively, with $r_1^- = 0$, $r_N^+ = R$ by definition and the kerf between successive rings being $r_p^- - r_{p-1}^+$. Relative time delays τ_p on each ring may also exist, allowing q_p to take the complex form

$$
q_p = \gamma_p + j\delta_p = |q_p| \, e^{j\theta_p} \tag{3}
$$

such that each ring emits pressure $q_p e^{-j\omega t} = |q_p| e^{j\theta_p} e^{-j\omega t}$ $= |q_p| e^{-j\omega(t-\tau_p)},$ where $|q_p|, \theta_p$, and τ_p are the respective ring magnitudes, phases, and time delays obtained from **(3)** as

$$
|q_p| = \sqrt{\gamma_p^2 + \delta_p^2}, \quad \theta_p = -j \ln(q_p / |q_p|), \quad \tau_p = \theta_p / \omega.
$$
\n(4)

111. USE OF 1-D FOURIER-BESSEL SERIES

A. Application of Fourier-Bessel Series

We begin by applying 1-D Fourier-Bessel series [l], **[2]** to model the quantized surface pressure $q(r)$ as an infinite set of known basis Bessel functions. This series is defined by

$$
q(r) = \sum_{i=1}^{\infty} A_i \cdot J_0(\alpha_i r)
$$

\n
$$
\alpha_i = x_i/a, \quad J_0(x_i) = 0
$$

\n
$$
A_i = \frac{2}{a^2 J_1^2(x_i)} \int_0^a q(r) \cdot J_0(\alpha_i r) r dr
$$
\n(5)

where $J_0(\cdot)$ is the Bessel function of the first kind of order zero, *Ai* is the appropriate set of weighting coefficients, and the roots x_i are the known infinite set of monotonically increasing positive solutions to $J_0(x_i) = 0$. The series in (5) is then valid over the radial range $0 \le r \le a$, where *a* is any desired modelling aperture subject to the constraint $q(a) = 0$, as $J_0(\alpha_i a) = 0$ by definition of $J_0(x_i) = 0$. (Note that for ease of discussion, we use the term aperture here to refer to the radial dimension *a* and not the corresponding diameter *Za).*

B. Interpretation of Fourier-Bessel Series

What the Fourier-Bessel series in *(5)* represents physically is that we may model both the quantized surface pressure $q(r)$ at $z = 0$ as $q_1 \ldots q_N$ over N rings of the transducer in the range $0 \le a \le R$, and then as $q(r) = 0$ beyond the transducer surface $a > R$, as the (relative) surface pressure is zero by definition beyond the transducer edge. This satisfies the condition $q(a) = 0$ as required by the definition of the series. Then, $q(r)$ in (5) equates to a surface profile $q(r)e^{-j\omega t} = \sum_{i=1}^{\infty} A_i \cdot J_0(\alpha_i r) \cdot e^{-j\omega t}$, and, if this profile is implemented over an infinite aperbeyond the transducer surface $a > R$, as the (relative) surface pressure is zero by definition beyond the transducer
edge. This satisfies the condition $q(a) = 0$ as required by
the definition of the series. Then, $q(r)$ in (diffracting Bessel beam solution 141, *[7]* at the transducer

surface $z = 0$ to the wave (1). The general propagating solution for $z > 0$ is then $A_i \cdot J_0(\alpha_i r) \cdot e^{j\beta_i z} \cdot e^{-j\omega t}$, and the total field $f(r, z, t)$ may therefore be defined as the sum of all of these components, namely

$$
f(r, z, t) = \lim_{a \to \infty} \sum_{i=1}^{\infty} A_i \cdot J_0(\alpha_i r) \cdot e^{j\beta_i z} \cdot e^{-j\omega t}
$$

$$
\beta_i = \sqrt{k^2 - \alpha_i^2}
$$
 (6)

in which $k = \omega/c$ is the wavenumber, the real parameter α_i \geq 0 is the scaling parameter in the *r* direction, and β_i is the component in the *t* direction. In addition, because the positive Bessel roots $x_i \approx \pi i - \pi/4$ in (5) increase monotonically with index *i*, the corresponding $\alpha_i = x_i/a$ terms in *(6)* also increase monotonically for a given valne of *a.* This causes a change in propagation characteristics for the distinct cases $\alpha_i \leq k$ and $\alpha_i > k$. Because the wavenumber k is assumed real, then β_i is purely real when $\alpha_i \leq k$ and all corresponding components $A_i \cdot J_0(\alpha_i r)$. $e^{j\beta_i z} \cdot e^{-j\omega t}$ propagate to infinity in the *z* direction for $z > 0$. However, when $\alpha_i > k$, the values β_i become purely imaginary, and these propagate as $A_i \cdot J_0(\alpha_i r) \cdot e^{-|\beta_i| z}$. $e^{-j\omega t}$, which is to say with exponential decay in the $z > 0$ direction. These are evanescent beam components, which are usually all negligible, as the magnitudes of $|\beta_i|$ involved typically cause the beam amplitudes $A_i \cdot J_0(\alpha_i r) \cdot e^{-|\beta_i|z}$ to decay to negligible levels within the first few wavelengths of the transducer surface.

Iv. THEORY FOR FIELD COMPUTATION

A. Computation Mechanism

Therefore, if we define $l(k, a)$ as the number of nonnegligible (generally exclusively nonevanescent) components for a given application, the original infinite sum in (6) effectively becomes replaced by the truncated finite sum

$$
f(r, z, t) = \lim_{\alpha \to \infty} \sum_{i=1}^{l(k, \alpha)} A_i \cdot J_0(\alpha_i r) \cdot e^{j\beta_i z} \cdot e^{-j\omega t}, \quad z > 0
$$
\n(7)

in which all required parameters A_i , α_i , β_i are now known from (5) and (6). Further, from the root approximation $x_i \approx \pi i - \pi/4$ and the definition $\alpha_i = x_i/a$, we have $\alpha_i a \approx \pi i - \pi/4$. Hence, the coefficient index $i = l(k, a)$ at which the swap bctween nonevanescent and evanescent characteristics occurs $(\alpha_i = k)$ is given by

$$
l(k,a) \approx ka/\pi + 1/4. \tag{8}
$$

Note that this is independent of the surface pressure profile *q(r),* but proportional to both wavenumber *k* and modcharacteristics occurs $(\alpha_i = k)$ is given by
 $l(k, a) \approx ka/\pi + 1/4.$ (8)

Note that this is independent of the surface pressure pro-

file $q(r)$, but proportional to both wavenumber k and mod-

elling aperture a. Therefore, im still need to sum an infinite number of terms to compute $f(r, z, t)$ exactly. Clearly, this is not possible from a practical point of view, but it is possible is to iterate *a* toward

lations $f(r, z, t)$ to converge to within an acceptable level.
Typically, we might use $a = 10R$, $a = 20R$, $a = 30R$, and so on, until the maximum change in field intensity between successive *a* values drops either to within a given number of decibels, or a given relative percentage (e.g., *0.1%).* See Section V for a detailed numerical example of computing the field.

B. Numerical Aspects

For annular arrays as per Section II-B, the integral for the weighting coefficients A_i in (5) may be evaluated piecewise over the N rings $p = 1, ..., N$ to give

$$
A_{i} = \sum_{p=1}^{N} C_{i,p} \cdot q_{p}
$$
\n
$$
C_{i,p} = 2 \left[r_{p}^{+} J_{1}(\alpha_{i} r_{p}^{+}) - r_{p}^{-} J_{1}(\alpha_{i} r_{p}^{-}) \right] / ax_{i} J_{1}^{2}(x_{i})
$$
\n(9)\nFig. 1. Quantization profile (upper) and\n
$$
C_{i,p} = 2 \left[r_{p}^{+} J_{1}(\alpha_{i} r_{p}^{+}) - r_{p}^{-} J_{1}(\alpha_{i} r_{p}^{-}) \right] / ax_{i} J_{1}^{2}(x_{i})
$$
\n(9)

where $J_1(\cdot)$ is the first-order Bessel function of the first kind. The coefficients A_i are, therefore, generally complex, as $C_{i,p}$ is real by definition, but q_p are complex according $q_{N+1} = 0$, inner radius $r_{N+1}^- = r_N^+ = R$, and outer rato Section II-B. Notice also that all coefficients $C_{i,v}$ are functions of geometry only, and need only be computed once for a given transducer layout, as they are independent of the ring pressures q_p . One may then also employ the approximations

$$
J_0(x) \approx \sqrt{2/\pi x} \cdot \cos(x - \pi/4), \quad x \ge 1
$$

$$
J_1(x) \approx \sqrt{2/\pi x} \cdot \cos(x - 3\pi/4), \quad x \ge 2
$$
 (10)

from [1] to obtain the roots x_i to $J_0(x_i) = 0$ as $x_i \approx \pi i$ - $\pi/4$, and, hence, $J_1(x_i) \approx \sqrt{2/\pi x_i}$, leading to $ax_i J_1^2(x_i) \approx$ $2a/\pi$ and hence

$$
C_{i,p} \approx \pi \left[r_p^+ J_1(\alpha_i r_p^+) - r_p^- J_1(\alpha_i r_p^-) \right] / a \tag{11}
$$

which shows that $|C_{i,p}|$ is bounded for all finite r_p^+ , r_p^- , a and all *i*, as $|J_1(\cdot)| < 0.6$ by definition. In fact, $|\tilde{C}_{i,p}| \to 0$ as $\alpha_i \to \infty$, because $\alpha_i \to \infty$ as $i \to \infty$ and $|J_1(x)| \to 0$ as $x \to \infty$ by definition. Therefore, $|A_i| \to 0$ as $\alpha_i \to \infty$ because all ring values q_p are fixed and finite. Note also two further numerical features. First, from (11) and (9) $|A_i|$ decreases with increasing *a*, as $|C_{i,p}|$ decreases with increasing *a* for a fixed value of *i.* Second, because the monotonically increasing Bessel root values x_i are given monotonically increasing Bessel root values x_i are given
by $x_i \approx \pi i - \pi/4$, then $x_i - x_{i-1} \approx \pi$, and, hence, consecutive alpha parameters in the series are spaced by an amount $\alpha_i - \alpha_{i-1} = (x_i - x_{i-1})/a \approx \pi/a$. This spacing decreases with increasing a , and an increasing number of nonevanescent alpha parameters appear in the nonevanesccnt region $\alpha_i \leq k$ as *a* is increased. It is also the reason amount $\alpha_i - \alpha_{i-1} = (x_i - x_{i-1})/a \approx \pi/a$. This spacing in
decreases with increasing *a*, and an increasing number of *c*
nonevanescent alpha parameters appear in the nonevanes-
cent region $\alpha_i \le k$ as *a* is increased. It i and an infinite number of α_i parameters appear in a finite interval $0 < \alpha_i \leq k$.

Notice also that implementing $a > R$ is equivalent to considering an equivalent transducer with $N' = N + 1$ rings, in which the outer ring has quantization level $q_{N'} =$

Fig. 1. Quantization profilc **(upper)** and **nonevanescent** Fourier-Bessel coefficients A_i (lower) for modelling aperture $a = R$.

dius $r_{N+1}^+ = a$. Theoretically, this gives rise to a new sum $A_i = \sum_{p=1}^{N'=N+1} C_{i,p} \cdot q_p$ in place of $A_i = \sum_{p=1}^{N} C_{i,p} \cdot q_p$ in (9), but because $q_{N'} = 0$ by definition, the sum remains fixed in practice at that given in (9). However, the $C_{i,p}$ coefficients change in (9) through their dependence on *a,* and, therefore, different A_i weighting distributions are obtained for different values of *a.* (See the plots supplied in Section V for a full illustration.)

V. EXAMPLE OF FIELD COMPUTATION

In this section, we illustrate the Fourier-Bessel method numerically and compare it with both experimental field results and Rayleigh-Sommerfeld simulations with Fresnel approximations.

A. Transducer Definition

Consider the Bessel transducer of Lu and Greenleaf described in [5]. The transducer is an $N = 10$ -ring Besseldesign transducer whose ring edges are located nominally at the first 10 zeros of $J_0(\alpha r)$, where $\alpha = 1202.45 \text{ m}^{-1}$. In practice, the transducer has kerf of approximately 0.2 mm, such that in terms of the notation of Section II-B, $r_1^- = 0$, $r_1^+ = x_1/\alpha - \text{kerf}/2, r_2^- = x_1/\alpha + \text{kerf}/2,$ and so on. Operating conditions are $f = 2.5$ MHz in water at speed of sound $c = 1500 \text{ m/s}$, giving wavenumber $k = 10471.98 \text{ m}^{-1}$. The $R = 25$ -mm transducer has its ring pressures q_p (solid lines) in Fig. 1, upper) chosen as the peak value of each respective Bessel lobe (dashed line in Fig. 1, upper).

B. Illustration of Field Convergence

The convergence principle for the field calculation as *B. Illustration of Field Convergence*
The convergence principle for the field calculation as
 $a \to \infty$ is illustrated in Fig. 1 through 5. Beginning with

Fig. 2. -30 dB contour levels for field calculation with $a = R$.

Fig. 3. Quantization profile (upper) and nonevanescent Fourier-**Bessel coefficients** A_i **(lower) for modelling aperture** $a = 5R$ **.**

 $a = R = 25$ mm, we obtain $l(k, a) = 83$ nonevanescent *A;* coefficients, as shown in Fig. 1 (lower). See **[16]** and **[SI** for a full discussion of the significance of the different *Ai* weightings and their associated field components. The calculation field based on $a = R$ is then shown in Fig. 2, where the lines shown represent the calculated -30 dB field contours. Note that we do not yet assume this to be the true field bccause we have not yet begun the iteration process of allowing $a \rightarrow \infty$. In Fig. 3 and 4, we then show the corresponding results for modelling aperture $a = 5R = 125$ mm. In this case, there are $l(k, a) = 416$ nonevanescent A_i coefficients as shown in Fig. **3** (lower). Notice that the number of coefficients has increased while the relative magnitudes have decreased as a has been increased; both these properties were predicted by the discussion given in Section IV-B.

Fig. 4. -30 dB contour levels for field calculation with $a = 5R$.

Fig. 5. -30 dB contour levels for field calculation with $a = 10R$.

The -30 dB field plot in Fig. 4 has also changed considerably with respect to Fig. 2, and so the modelling aperture is then increased to $a = 10R = 250$ mm for which $I(k, a) = 833$, and the corresponding field plot is given in Fig. *5.* This plot has changed much less than **u'as** the case previously, indicating that the convergence of the field calculation has begun to takc place. As the aperture is increased to $a = 20R$ and $a = 30R$, no further visible changes are apparent in the -30 dB plots, and the maximum relative change in field intensity encountered anywhere in the entire region of interest is found to drop to within less than **1%.** For practical purposes, we consider changes are apparent in the -30 dB plots, and the max-
imum relative change in field intensity encountered any-
where in the entire region of interest is found to drop to
within less than 1%. For practical purposes, we convergence to have taken place at $a = 30R = 750$ mm for which $l(k, a) = 2500$ coefficients. (See Table I for a fuller set of field parameters as a function of modeling aperture

Number of **miievanescent** coefficicnts

Maximum relative field intensity change present *(W)*

TABLE **I** 'FIELD CALCULATIoN 'PARAMErERS **AS A** FUNCI'ION **OF** INCREASlXG **APERTURE** RATIO.

Aperturc ratio $a=R$ $a=5R$ $a=10R$ $a=20R$ $a=30R$

Fig. 6. Nonevanescent coefficient values A_i versus alpha values α_i for $a = R$, $a = 5R$, $a = 10R$ and wavenumber $k = 10471.98$ m⁻¹.

Alpha value *ai*

0

<--0.05 O1.. *0* 2000 **4000** *6000 8000* 10000

a). Finally, in Fig. 6 and **7:** respectively, the *Ai* coefficients are plotted against α_i and corresponding axicon angles $\zeta_i = \sin^{-1}(\alpha_i/k)$ for the three first apertures $a = R$. $a = 5R$, $a = 10R$. As commented previously, the coefficient magnitudes decrease as the number of nonevanescent coefficients increasc with increasing *u.* However, what appears to remain constant is the shape (although not magnitude) of the envelope of the field component distributions, sugmagnitudes decrease as the number of nonevanescent coefficients increase with increasing a. However, what appears
to remain constant is the shape (although not magnitude)
of the envelope of the field component distributio marily around particular values of α and corresponding axicon angles ζ . Moreover, these principle values appear to be the principal limited diffraction field components already analyzed recently in [16].

C. Cornpanson *with Experimental und* $Rayleigh-Sommerfeld$ Results

Fig. $8(a)$ shows the greyscale image of the transducer field evaluated with $a = 30R$. Next to it in Fig. 8(b), we see the excellent agreement with the experimental field result of [5]. Fig. 8(c) then shows the Rayleigh-Sommerfeld field computation without Fresnel approximation, which again agrees very closely with the Fourier-Bessel calculation. Finally, in Fig. $8(d)$, we show the Rayleigh-Sommerfeld computation without Fresnel approximation, which suffers from errors in the very nearficld. Notice that these errors

Comparision of A_i Coefficients for $a=R$, $a=5R$, $a=10R$

83 416 833 1666 2500 $\frac{a-3a}{83}$ $\frac{416}{16}$ $\frac{833}{83}$ $\frac{1666}{15.69}$ $\frac{2500}{1.93}$ 0.79

 -84.41 -72.34 -74.32 -73.44

Fig. 7. Nonevanescent coefficient values A_i versus axicon angles $\zeta_i =$ $\sin^{-1}(\alpha_i/k)$ for $a = R$, $a = 5R$, $a = 10R$ and wavenumber $k =$ 10471.98 m^{-1}

arc absent in the Fourier-Bessel calculation. Im addition to its accuracy, the Fourier-Bessel algorithm ran 14.1 and 3844.3 times faster than the Rayleigh-Sommerfeld algorithms with and without Fresnel approximations, respectively (see [Table 11\)](#page-6-0).

VI. THEORY FOR TUNING

$A.$ *Computation Mechanism*

The theory for tuning of annular arrays using Fourier-Bessel series has already been outlined in **[15]** in the context of tuning a Bessel beam on an equal area annular array. Here, we first review it and then apply it more extensively than previously for a different type of array in Section VII. The field $f(r, z, t)$ in (7) may be separated into the product of time component $e^{-j\omega t}$ and spatial component $f(r, z)$ in terms of the ring pressures $q_p = \gamma_p + j\delta_p$ by combining (7) , (9) , and (3) to obtain

$$
f(r, z, t) = e^{-j\omega t} \cdot \lim_{a \to \infty} \sum_{i=1}^{l(k,a)} J_0\left(\frac{x_i r}{a}\right) \times \left[\sum_{p=1}^N C_{i,p} \cdot e^{j\beta_i z} \cdot (\gamma_p + j\delta_p)\right] = e^{-j\omega t} \cdot f(r, z) \quad (12)
$$

Fig. 8. Field for quantized CW Bessel beam. **a)** Field **calculsted** by Fourier-Bessel theory. b) Experimental ficld. **c)** Field **calculated** by Rayleigh-Sommerfeld formula without Fresnel approximation. d) Field calculated by Rayleigh-Sommerfeld formula with Fresnel approximation.

TABLE **11** COMPARISON OF RELATIVE RUN TIMES FOR DIFFERENT FIELD SIMULATION METHODS

Simulation method	Fourier-Bessel	Rayleigh-Sommerfeld with Fresnel	Rayleigh-Sommerfeld without Fresnel
Time (h:min:s)	00:00:28	00:06:35	29:54:00
Time (s)	28	395	107640
Time (relative)		14.1	3844.3

The spatial field component $f(r, z)$ postmultiplying $e^{-j\omega t}$, therefore, has complex form $f(r, z) = f^{\Re}(r, z) + j f^{\Im}(r, z)$, where $f^{\Re}(r, z)$ and $f^{\Im}(r, z)$ are the real and imaginary components, respectively. From this, the field at all points of interest $r = r_u$, $z = z_v$ $(u = 1...n_u$ and $v = 1...n_v)$ may be written as (13) (see next page) in which $M_{i,p,v}^{\mathcal{R}}$ and $M_{i,p,v}^{\mathcal{S}}$ are the real and imaginary parts of the product $M_{i,p,v} = C_{i,p} \cdot e^{j\beta_i z_v}$, respectively. Eq. (13) then has block form

$$
F = MQ \tag{14}
$$

in which

$$
Q = [\gamma_1, \delta_1, \dots, \gamma_N, \delta_N]'
$$
 (15)

on the right-hand side is the vector containing the reel and imaginary parts of all N adjustable ring pressures q_1, \ldots, q_N . This vector has dimension $Q = Q\{2N, 1\},\$ where the notation Q {rows, cols} indicates the number of rows and columns, respectively.

We then aim to minimize the envelope of the difference in field intensity between the actual field vector $F = F{2n_un_v, 1}$ and some desired field vector $D =$ $D\{2n_{u}n_{v},1\}$ in a least squares sense by adjusting all components $\gamma_1, \delta_1, \ldots, \gamma_N, \delta_N$ from the ring pressures in (3) appropriately. To achieve this, first define the desired field $d(r, z, t) = e^{-j\omega t} \cdot d(r, z)$ as a product of time component $e^{-j\omega t}$ and spatial component $d(r, z)$. The real and imaginary spatial components $d^{\Re}(r_u, z_v)$, $d^{\Im}(r_u, z_v)$ may then be written out at all points of interest $(u = 1...n_u)$ and $v = 1 \ldots n_v$) in the same format as the vector *F*, i.e. (16) (see next page) and the least squares minimization problem with respect to Q then reduces to the minimization of the error sum $S = [F - D]'[F - D]$. Substituting $F = MQ$

$$
\begin{bmatrix}\nf^{\Re}(r_1, z_1) \\
f^{\Im}(r_1, z_1) \\
\vdots \\
f^{\Re}(r_u, z_v) \\
f^{\Im}(r_u, z_v) \\
\vdots \\
f^{\Re}(r_u, z_v)\n\end{bmatrix} = \n\begin{bmatrix}\n\lim_{a \to \infty} \sum_{i=1}^{l(k,a)} J_0\left(\frac{x, r}{a}\right) \left[\begin{array}{r} + M_{i,1,1}^{\Re}, -M_{i,1,1}^{\Re}, \dots, + M_{i,N,1}^{\Re}, -M_{i,N,1}^{\Re} \end{array} \right] \\
\lim_{a \to \infty} \sum_{i=1}^{l(k,a)} J_0\left(\frac{x_i r_u}{a}\right) \left[\begin{array}{r} + M_{i,1,v}^{\Re}, -M_{i,1,v}^{\Re}, \dots, + M_{i,N,v}^{\Re}, -M_{i,N,v}^{\Re} \end{array} \right] \\
\lim_{b \to \infty} \sum_{i=1}^{l(k,a)} J_0\left(\frac{x_i r_u}{a}\right) \left[\begin{array}{r} + M_{i,1,v}^{\Re}, -M_{i,1,v}^{\Re}, \dots, + M_{i,N,v}^{\Re}, -M_{i,N,v}^{\Re} \end{array} \right] \\
\lim_{b \to \infty} \sum_{i=1}^{l(k,a)} J_0\left(\frac{x_i r_{n_u}}{a}\right) \left[\begin{array}{r} + M_{i,1,v}^{\Re}, -M_{i,1,v}^{\Re}, \dots, + M_{i,N,v}^{\Re}, -M_{i,N,v}^{\Re} \end{array} \right] \\
D = \left[d^{\Re}(r_1, z_1), d^{\Im}(r_1, z_1), \dots, d^{\Re}(r_u, z_v), d^{\Im}(r_u, z_v), \dots, d^{\Re}(r_{n_u}, z_{n_v}), d^{\Im}(r_{n_u}, z_{n_v}) \right],\n\end{bmatrix} (16)
$$

from (14) and minimizing with respcct to *Q* gives least squares solution $Q = Q_{ls} \{2N, 1\}$ as

$$
Q_{ls} = [M'M]^{-1}M' \cdot D. \tag{17}
$$

Note that this tuning method has some similarities to the previous limited diffraction design of [17], hut possesses two major differences in approach and objective. The first is that [17] considers a single value of α and then combines weightings of $J_n(\alpha)$ for different Bessel orders *n*, whereas this method considers multiple values of α_i but combines weightings of $J_n(\alpha_i)$ for $n = 0$ only. The second is that [17] derives weighting parameters on the assumption that the resulting Bessel functions could be realized perfectly in practice on the transducer surface, whereas this method considers specifically the effects of quantization on the transducer and derives a tuning scheme that takes thcse restrictions into account.

B. Numerical *Aspects*

The vector dimensions $F = F{2n_un_v, 1}$ and $Q =$ *Q{2N:* 1) cause the large matrix *M* premultiplying Q to have dimension $M = M\{2n_{u}n_{v}, 2N\}$ and the inverse $[M'M]^{-1}$ in (17) must exist for the solution to be realizable. Hence, $M'M = M'M\{2N,2N\}$ must have full rank 2N, and this imposes a requirement of $n_u n_v \geq N$ to prevent *M* (and thereby *M'M)* from being rank deficient for dimensional reasons. In addition, consideration needs to be given to spatial sampling rates. From **(7),** component *i* of the sum propagates in the *z* direction as $e^{j\beta_i z}$ with wavelength $2\pi/\beta_i$ for the nonevanescent components. The shortest possible wavelength is, therefore, that corrcsponding to the maximum possible nonevanescent value of β_i , namely $\beta_{\text{max}} = k$ when $\alpha_i = 0$ in (6). This gives a wavelength of $2\pi/k$, which, to comply with the Shannon sarnpling theorem, dictates a sampling interval in the *t* direction of π/k or lower. In the radial direction r, the approximation $J_0(\alpha_i r) \approx \sqrt{2/\pi \alpha_i r} \cdot \cos(\alpha_i r - \pi/4)$ from (10) allows us to approximate the radial oscillations as a cosine

function of wavelength $2\pi/\alpha_i$. The minimum wavelength possible is then also $2\pi/k$, corresponding to the maximum nonevanescent value $\alpha_i = k$ possible in (6). Therefore, this also leads to a sampling interval of π/k or lower in the r direction. Finally, we also need, in practice, to iterate for different values of *M* for $a \to \infty$, calculate Q_{ls} for each value of *a,* and wait for the corresponding quantization magnitudes $|q_p|$ and phases θ_p or time delays τ_p from (4) to converge to within acceptable levels. (See Section VI1 for a full numerical example).

VII. **EXAMPLES** OF TUNING

For tuning examples, we consider the focusing of a Gaussian beam with focal length $F = 120$ mm on the given annular array. This is obtained initially without any least squares tuning by quantizing both the ring amplitudes $|q_p|$ and phases θ_p (3) over each annulus according to

$$
q_p = e^{-r_p^2/\sigma^2} \cdot e^{jk(F - \sqrt{F^2 + r_p^2})} = |q_p| \, e^{j\theta_p},\qquad(18)
$$

which is a discrctizcd version of the Gaussian surface pressure expression given in [5] in which $r_1 = 0$, $r_p =$ $(r_p^- + r_p^+)/2$, $(p = 2...10)$, $\sigma = 15$ mm, $F = 120$ mm, and $k = 10.4798$ mm⁻¹. Eq. (18) gives amplitudes and phases as per rows 1 and 2 in Table 111, for which all *Ai* coefficients are imaginary, as all q_p are imaginary.

A. Experimental *Gaussiun* Field

The Fourier-Bessel ficld calcnlation with *a* = *30R* for (18) leads to the field in Fig. 9(a) over the intervals $0 \le r \le 24$ mm and $5 \le z \le 210$ mm. The result demonstrates a focus around $z = 120$ mm as expected, along with some nearfield diffraction in the region of $z < 100$ mm. We now compare this with the experimental result obtained previously in $[5]$, whereby the array was quantized with the same discrete magnitudes $|q_p| = e^{-r_p^2/\sigma^2}$ as per (18) but, in

TABLE I11 QUANTIZATION AMPLITUDES AND PHASES FOR PANELS IN FIG. **9(A)** AXD **10(A** AND **U).**

	Ring Number									
	$p=1$	$p=2$	$p=3$	$p=4$	$p=5$	$p=6$	$p=7$	$p = 8$	$p = 9$	$p=10$
$ q_p $ Fig. 9(a)	1.000	0.953	0.857	0.726	0.578	0.434	0.306	0.203	0.127	0.079
θ_p/π Fig. 9(a)	$\mathbf{0}$	-0.151	-0.482	-1.003	-1.711	-2.607	-3.689	-4.955	-6.404	-7.910
$ q_p $ Fig. 10(a)	1.001	-0.956	0.860	0.726	0.578	0.434	0.306	0.204	0.127	0.079
θ_p/π Fig. 10(a)	-0.001	-0.151	-0.482	0.998	0.289	-0.607	0.311	-0.955	-0.404 .	0.090
Unwrapped				$0.998 - 2$	$0.289 - 2$	$-0.607 - 2$	$0.311 - 4$	$-0.955 - 4$	$-0.404 - 6$	$0.090 - 8$
$\theta_p/\pi - 2*int$				$=-1.002$	$=-1.711$	$=-2.607$	$=$ -3.689	$=-4.955$	$=-6.404$	$=-7.910$
$ q_p $ Fig. 10(b)	1.007	0.959	0.862	0.726	0.578	0.434	0.305	0.202	0.126	0.077
θ_p/π Fig. 10(b)	-0.001	-0.152	-0.482	0.998	0.289	-0.607	-0.311	-0.955	-0.404	0.092
Unwrapped				$0.998 - 2$	$0.289 - 2$	$-0.607 - 2$	$0.311 - 4$	$-0.955 - 4$	$-0.404 - 6$	$0.092 - 8$
$\theta_p/\pi-2*int$				$=-1.002$	$=-1.711$	$=-2.607$	$=-3.689$	$=-4.955$	$=-6.404$	$=-7.908$

Fig. 9. Tuning **results** for **a** CW focused Gaussian beam. **a)** Focused **Gaussian beam** field calculated with Fourier-Bessel theory. b) **Erperimental** tuning using an **acoustic** lens. c) Raylcigb-Sornmerfeld verification of *(a).* d) Rayleigh-Sommerfeld **verification** *of* (b).

practice, with continuous phases $\theta(r) = k(F - \sqrt{F^2 + r^2})$ obtained by placing an acoustic lcns across the surface of the transducer (see [5] for details). The resulting experimental field is given in Fig. $9(b)$, demonstrating an almost identical focus to Fig. $9(a)$, but this time without the diffraction for $z < 100$ mm. This difference appears to be due to the difference between the quantized phases in Fig. $9(a)$ and the continuous phases in Fig. $9(b)$. To verify that this is the case as opposed to any sinnilation error in the Fourier-Bessel method for the casc of complex ring pressures, we then also simulate the ficlds for both Fig. 9(a and b) using the Rayleigh-Sommerfeld diffraction formula. The results are given in Fig. $9(c \text{ and } d)$, respectively,

demonstrating that both Fig. 9(a and b) are indeed consistent with the independent Rayleigh-Sommerfeld method; therefore, the Fourier-Bessel sirnulation method for complcx ring prcssurcs still holds good.

B. Venfication of Tunjng Algorithm Using Gaussian Field

Therefore, with full confidcncc in the Fourier-Bessel field evaluation method and taking Fig. 9(a) **as** the benclimark for what we know may be achieved given the constraints of discretized quantization phases, we now consider how the array could be tuned using that desired field but in the absence of the underlying (18). First, we take

Fig. 10. Tuning results for a CW focused Gaussian beam. (a) Tuning result using the whole field Fig. 9(a). b) Tuning result using a 48-mm
vertical line of the field in Fig. 9(a) through the focal distance $z = F = 120$ mm. c **as** (b) but with vertical **line** of 15 mm.

the entire field in Fig. 9(a) to construct the desired field vector D in (16), (17), and apply the tuning algorithm of Section VI with lateral (r) and axial *(z)* resolutions of 0.3 mm each. (Notice that these comply with the required sampling intervals of $\pi/k = 0.30$ mm or lower derived in Section VI. If the tuning theory is correct, we should then expect to retrieve the exact quantization magnitudes and phases that generated this field, as the tuning is recreating a field that we already know. In practice, wc do indeed find this to be the case, with the least squares ring pressures and phases (rows **3** and **4** of Table **111)** being virtually identical to the original phases (rows 1 and 2 of Table III). The field for the subsequent least squares ring pressures is given in Fig. $10(a)$; compare this with the original given field in Fig. 9(a). Hence, we have now established that both the field calculation and tuning methods are valid and accurate.

$C.$ *Tuning Using Reduced Gaussian Field*

A drawback, however, is that the tuning algorithm takes a long time to run when using the entire field data (approximately 6 h on a 600-MHz Pentium III PC programmed in C). So we now investigate how well it performs when substituting a much smaller slice of field data for the desired vector *D.* To do this, take only a single line of the field data sampled across the 48-mm radial cross-section of interest $(0 \le r \le 24$ mm) in the plots at the focal distance $z = 120$ mm. The computation time is then reduced to only a few minutes while the ring pressures are still found to be retrieved satisfactorily [sec rows 5 and 6 of Table 111 and the corresponding field plot in Fig. 10(b). We then attempt to reduce the data set even further by sampling first only the first 20 mm ($0 \le r \le 10$ mm) and second only the first 15 mm ($0 \le r \le 7.5$ mm) symmetrically around the central axis $r = 0$ at $z = 120$ mm from Fig. 9(a) [see Fig. 10(c and d); respectively]. Here; we see a degradation in the desired field because *of* the appearance of unwanted sidelobes on either side of the central focal zone. The reason for this is that, in specifying sampling regions of only $0 \le r \le 10$ mm and $0 \le r \le 7.5$ mm, respectively, the algorithm is not concerncd with what happens beyond these boundaries, and, thercfore, the field is not constrained in any way in those areas. This brings to light the point that one must be careful to specify a large enough area of interest for the least squares algorithm, if one is to be sure of obtaining a satisfactory field in the entire region of practical usage. Clearly, on the one hand, one is interested in keeping the algorithm data to a minimum from the point of view of computation time, but, on the other hand, one must ensure enough data to avoid degradation of the field in other areas that may still be of interest. A suitable global design to balance these objectives has not yet been investigated, and some degree of trial and error is needed at present to arrive at a satisfactory conclusion.

Fig. 11. Timing **results** [or simple desired fields. **a)** Tuning using **a** \ertical **line** through the **focus** at *z* = 120 mm. The line has **a** value of zero except in the central portion of a height of 2.54 mm, whose value is e^{jkz} (see upper left-hand corner of the figure). b) Tuning using the **Sam line as** in *(8) except* that **the** width is 10 mm centered at *z* = **120** mm. c) The same **as (a) except** that the width is **24.4** mm. d) The same as (a) with width of 240 mm.

D. Tuning *Using* Generalized Field

Finally, we now also assume that the pre-computed discretized Gaussian field of Fig. 9(a) is not available in practice and consider the more practical problem of tuning based on a simpler design rnlc. For this, wc design a field with a stepwise amplitude distribution based around $r = 0, z = 120$ mm, in the form of a vertical bar (with respect to the field plots) as shown in the upper left-hand corner of Fig. 11. The center of the bar is located at the point $r = 0$, $z = 120$ mm defined as the focus of the desired field and the total length (height) of the bar is 48 mm ($0 \le r \le 24$ mm) as per the tuning example of Fig. 10(b) previously. The height of the central section $0 \le r \le 1.27$ mm with distribution e^{jkz} is 2.54 mm (fullwidth at half-maximum lateral resolution *GR_L* in [5]), and the width of bar is changeablc while the field outside the range $0 \le r \le 1.27$ mm is defined as zero for 1.27 mm $\langle r \leq 24$ mm. Fig. 11(a) is then the tuning result for a single vertical line sampled at $z = 120$ mm, and Fig. $11(b \text{ and } c)$ are for bars of widths 10 mm (b) and 24.4 mm (c) (depth of field GFz_{max} in [5]). These represent a gradual lengthening of the desired depth of focus between each plot, with this being extended to a total width of 240 mm (double the focal length) in (d). In this latter case, **we** observe that the extension of thc focal zone is achieved at the cost of a significant increase in both the width of the mainlobe and the magnitude of supporting sidelobe Icvels. See Table IV for quantization amplitudes and phases corresponding to the four panels in Fig. **11.** Finally. Fig. **12** shows the convergence of least squares quantization magnitudes $|q_p|$ and phases θ_p as a function of increasing aperture ratio a/R for the tuning example of Fig. $11(a)$.

VIII. CONCLUSIONS **AND** FURTHER WORK

We have discussed a method for computing and tuning the linear lossless field of flat annular arrays using 1-D Fourier-Besscl series. The series corresponds to a set. of Bessel beams propagating into the medium, which, in the limit, provides a linear mapping between the ring pressures on the transducer surface and the field at any point in space. The Fourier-Bcssel field calculation method was found to be both quicker and more accurate close to the transduccr surface than the Rayleigh-Sommerfeld method with Fresnel approximation when applied to a 10-ring annular array operating at 2.5 MHz in water. The tuning method allowed us to tune the field in different manners by defining different desired fields. However, some degree of trial and error was found to he necessary when employing different design criteria, and this is an area that could benefit from further investigation. It may also be that op timization schemes other than least squares could provide

Fig. 12. Convergence of quantization magnitudes $|q_p|$ and phases θ_p as a function of aperture ratio a/R for tuning example in Fig. 11(b).

benefits. There remain also several other extensions to the current method that require investigation. First, extension of the analysis to pulse wave (PW) fields both for field computation and tuning. Second, the current analysis is limited to annular arrays and its development of non-annular arrays both in CW and PW cases is of interest for more widespread application. This will require the use of 2-D Fourier-Bessel series, which are capable of modelling quantization profiles and fields that are non-circularsymmetric around the propagation axis. Some initial work has already begun on this, but there are many computational aspects still to be investigated; these are partly due to the increased number of transducer elements in such arrays and partly due to the fact that 2-D Fourier-Bessel series require the computation of 2-D Fourier-Bessel coefficients. In principle, the analysis for both field computation and tuning of **2-D** series follow the same general structure as in the I-D case. However, it may be that the increased numerical complexity of such schemes provide difficulty in implementation because of computational time, memory, or numerical stability requirements. An extension of all of these methods to a suitable model for lossy media is also of interest.

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Paul Fox was born in Sheffield, England to an English father and French mother. In 1987 he completed a B.A. Hons Engineering Scicncc degree at the University of Oxford, England. During in 1988 he was Research **Assis**tant at the Department of Aeronautical Engineering, University of Sydney, Australia, and later became Research Assistant at the Department of Cybernetics, University of Reading. England from 1988-1991. From 1991- 1995 **he** remained there to complete **a** part time Ph.D. in mathematical modeling and sig-

nal processing, before becoming Research Fellow in control theory and system identification at the University **of** Warwick, England from 1995-1998. He then moved to Norway from 1998-2000 as a European Union Marie Curie **Research** Fellow to work with Professor Sverre Holm on limited diffraction beams at the ultrasound at the Department of Informatics, University of Oslo. Following this he spent the last four months of 2000 as Postdoctoral Research Fellow in limited diffraction beams with Professor Jian-yu Lu at the Ultrasound Laboratory, Department of Biocginccring, University of Toledo, Ohio, USA. More recently he has **been a** postdoctoral researcher in ultrasound at the **Sensors** Lab, Department of Engineering Science, University of Oxford, UK from July 2001 to April 2002. As of May 2002 he became postdoctoral researcher in medical ultrasound at Orstcd. DTU, Technical University of Denmark. His current main intcrests **are** in field propagation and beamforming with emphasis in medical **applicatiuns.**

Jiqi **Cheng** was born in **Jiangsu,** China, in 1975. He received the B.S. degree and the M.S. degree in biomedical engineering in 1998 and 2000 respectively, both from Southeast University, Nanjing, China. He is currently pursing his doctorate degree in the Depart-
ment of Bioengineering at the University of

His research interests in ultrasound include Foledo, OH.
His research interests in ultrasound include
field calculation, beam forming, transducer
modeling, and medical imaging.

Jian-yu Lu (S'86-M'88-SM'99) is currently **a** professor in the Department of Biuengineering at the University of Toledo, Toledo, OH, and **an** adjunct professor of medicine at the Mcdical College of Ohio, Toledo, OH. Hc has **bccn** the Graduate Director of the Department of Bioengineering at thc University of Tolcdo since 1999. Before joining the University of Toledo as a professor in September 1997, he was an associate professor of biophysics at the Mayo Medical School and an associate consultant at the Department of Phys-

iology and Biophysics, Mayo Clinic/Fonndation. Rochester, MN. His research interests **arc** in acoustic imaging and tissue idcrrtification, medical ultrasonic transducers, and ultrasonic beam forming and propagation.

Dr. Lu has published rnariy **papers** in peer-reviewed journals. Two of his papers published in the IEEE Transactions on the Ultrasonics, Ferroelectrics, and Frequency Control (UFFC) in 1992 have received the Outstanding Paper Award from the UFFC society for the discov**ery** of X **waves** that, in theory, **can** propagate to an infinitc distance without spreading (diffraction-free). These waves have potential **ap** plications in medical imaging (in both ultrasound and optics).

Dr. **Lu** has received the Edward C. Kendall **(a** Nobel Laureate at Mayo Clinic) Award from the Mayo Alumni Association, **Mayo** Foundation, in 1992, for his meritorious research; the FIRST Award from the National Institutes of Health (NIH) in 1991; and the Biomedical Engineering Research Grant Award from the Whitaker Foundation in 1991, in addition to other long-term RO1 type of NIH grant award.

Dr. Lu **is** the Editor-in-Chief of the *IEEE Thnsactions* on *Ultmonics, Femdectncs.* **and** *Prequency Control.* **He was lhe** Technical Program Chair of tho 2001 IEEE International Ultraionics Symposium-a joint meeting with the World Congress on Ultrasonics held in Atlanta, GA, in October 2001. He has been an Exhibition Chair of the IEEE Ultrasonics Symposia for many **years** and is a member of the Teclrnicai Prograin Committee of Group **I** (Medical Ultrasonics). He serves in both the UFFC Web Committee and the UFFC Ultrasonics Committee. Dr. Lu is a senior member of the IEEE UFFC Society and a senior member of the American Institute of Ultrasound in Medicine (AIUM). He is also a managing editor of the online journal Frontiers in Bioscience.