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## IMPROVEMENTS IN MONITOR LOUDSPEAKER SYSTEMS

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As the recording industry enjoys the benefits of both digital and advanced analog recording technology, attention is appropriately focused on the use of compression driver and horn designs which are some 25 to 30 years old. Evolutionary improvements in woofers, compression drivers, and dividing networks combined with new constant coverage horn designs have resulted in frequency response more consistently uniform at all coverage angles (yielding flat power response) along with lowered distortion and increased acoustic power output at the frequency extremes.

### INTRODUCTION

High-quality studio monitor loudspeaker systems have evolved out of the theater traditions of the 1930's and 1940's. These systems, whether of 2-, 3- or 4-way design, invariably make use of compression drivers in their mid- and high-frequency sections for greater reliability at elevated output levels. These design traditions have been eschewed by the "audiophile" segment of the consumer market, who have in general preferred the relative smoothness and low distortion (at moderate levels) of cone and dome direct radiating systems. As audiophile record productions take on a more conspicuous profile, and as digital recording technology promises higher orders of performance in the studio, we once again examine and attempt to reconcile the apparent differences between that which the dedicated audiophile feels to be a state-of-the-art approach to loudspeaker design -- and that which the experienced recording engineer requires for his specific needs.

Among the chief performance parameters we have identified are uniform polar response and directivity, smooth power response and low distortion. A secondary requirement is for accurate stereophonic imaging at close-in listening positions in the studio control room. A new family of constant directivity horns has formed the basis of a new approach to monitor design, and we will now describe two monitor loudspeakers embodying them.

### THE CONSTANT DIRECTIVITY HORN

Horns used in previous monitors were of the radial type or, alternatively, straight exponential with a horizontally divergent acoustic lens. Either type offered wide but uncontrolled horizontal response at the expense of narrow vertical response at the highest frequencies and constantly rising directivity with frequency. Recent developments in horns (Keele [1], Henricksen and Ureda [2]) yield very uniform vertical and horizontal coverage patterns which change little with frequency. These horns provide surprisingly

constant angular coverage along with very stable directivity over their operating range. This alone cures the worst complaints about previous horns and makes systems using such horns truly a new generation of monitors.

Optimal horn parameters for monitor use are as follows.

- (1) Constant coverage angle with consistent polar patterns, both horizontally and vertically controlled over the total operating range (1000 Hz to 16,000 Hz).
- (2) Coverage angles wide enough to mate at crossover with a cone woofer (90° to 100° square or a Directivity Index [DI] of about 8 dB).
- (3) Faster flare than previously used, for lower second-harmonic distortion.
- (4) Shorter length to place woofer and horn in the same acoustic plane.

The first version of a constant directivity horn (Keele [1]) was basically a much improved radial horn with end flaring to combat midrange narrowing and maintenance of high frequency beamwidth by elimination of the typical radial horn neck. A second version described by Henricksen and Ureda [2] brought vertical angle control down to a lower frequency by flipping conventional horns 90° onto their sides, which allowed a much larger vertical height without the width growing too excessively. Flat surface flares with hard transitions were used for manufacturing ease. A third version by Keele combines the better points of the previous two with several improvements: The vertical flare is fed by a diffraction slot, which can be made narrow enough to feed a wide horizontal angle up to any desired frequency. The primary and end flares have rounded transitions as on the previous Keele horn, but rather than being arbitrarily rounded the side contours are defined by a three-term mathematical expression:

$$y = a + bx + cx^n$$

where  $b$  determines the initial wall angle and the  $cx^n$  term determines the amount of mouth flaring (see Fig. 1). The performance of such a horn is compared to a prior art radial horn and exponential horn with acoustic lens in Figure 2. Its -6 dB coverage angles and directivity index are quite consistent.

#### THE COMPRESSION DRIVER AND ITS EQUALIZATION

Previous horns were judged on their ability to generate flat response on axis with typical compression drivers. This persisted, even though compression drivers were known to fall off in power response above the midband. A typical driver whose power response rolls off at 6 dB per octave above 3 kHz (Newman [3]) would require a horn with a reciprocal increase in directivity index. The compression driver would then be acoustically equalized, but only on axis, as shown in Figure 3.

When this compression driver is loaded by a constant directivity horn, the axial response follows the power response of the driver, as shown by Figure 4. The very high midband efficiency given by horn loading allows passive equalization for both flat axial, and at the same time, flat power response. A typical network configuration is shown in Figure 5. The midband sensitivity is reduced

by the L-pad and dividing network. The highs are shunted across this attenuation by a second order bandpass filter tuned to the highest operating frequency. Independent control of the midrange and high end are an added benefit.

When the high end of the horn/driver combination is properly equalized (Figure 6,a), we can return the compression driver to the constant impedance plane wave tube and verify that its power response is now flat (Figure 6,b).

#### THE WOOFER

Woofers for monitor use have been slowly but constantly evolving over the past 20 years. The ideal woofer must have:

- (1) A smooth response curve with the required midband sensitivity.
- (2) Controlled directional characteristics.
- (3) High output at low distortion levels.
- (4) Freedom from dynamic offset problems.

It is commonly thought that 38 cm (15 inch) woofers cannot be used up to 1000 Hz due to poor frequency response and ragged polar characteristics. Fortunately, this need not always be the case. Figure 7 shows the beamwidth and directivity index of a 38 cm woofer mounted in an enclosure of typical size (0.17 cu m or 6 cu ft). The directivity rises smoothly, until at 1000 Hz it becomes an even match for a 100° by 100° horn.

The high price of Alnico V magnets has forced most manufacturers to use ferrite magnet structures, even though these were previously regarded as higher in distortion. Improved geometry and the use of flux modulation cancelling rings has reduced their distortion levels to less than the equivalent Alnico structure (Gander [4], Gilliom [5]). An added benefit is the elimination of the Alnico structure's tendency to demagnetize itself under high-power, low-frequency pulses.

Woofer dynamic offset is a problem long known about but seldom discussed or treated. With high input power at low frequencies, many woofers tend to shift their mean displacement forward or backward until the coil is nearly out of the gap. This is most likely to happen just above each low frequency impedance peak of a system. The result is a high level of second harmonic distortion and subjectively a bass character that loses its tightness at high acoustical output levels [4]. The cure for offset, as shown by T. H. Wiik [6], is a restoring spring force that increases in stiffness at high displacement in an amount that counterbalances the reduced B field at the extremes of voice coil travel. Such a nonlinear spider will in fact reduce distortion and eliminate the tendency to offset.

#### DIVIDING NETWORK CONSIDERATIONS

The main goal of good network design is to produce the flattest response over the widest range of listening angles. The three-dimensional position dependence of the frequency response of a loudspeaker system, as controlled by

the network, is generally overlooked. Response at off-angles and total power response are of major concern. Each trial network with good axial curves must be measured at many angles, up and down, left and right. Those that pass this phase of testing will of course be further tested for power response and directivity characteristics, and then be taken to the studio for measurement and exhaustive listening tests.

Crossover nulls that appear at off-axis angles are in inevitable consequence of the finite driver spacing. In non-coaxial designs the spacing is usually in the vertical plane, and it causes the woofer to listener and tweeter to listener distances to vary as the system axis is tilted. Linkwitz [7] shows that the angle between nulls is roughly defined by the wavelength of sound at the crossover frequency and the vertical spacing and is given by:

$$\alpha = \arcsin \left( \frac{\lambda_c}{2 d_1} \right),$$

where  $\alpha$  = half angle between nulls,

$\lambda_c$  = wavelength at crossover frequency, and

$d_1$  = center to center spacing (vertical array assumed).

For example, a 1000 Hz crossover frequency and a spacing of .4 meter yields:

$$d_1 = 0.4\text{m (16 in)},$$

$$\lambda_c = 0.34\text{m (13.5 in)},$$

$$\alpha = 25^\circ, \text{ or}$$

$$2\alpha = 50^\circ \text{ (arc between nulls)}.$$

The arc between nulls can be made more useful if it is tilted upward. That is, floor standing systems should be optimized for response at angles on axis and above. If the system is to be used above ear level then inverting it will once again yield the greatest latitude of listener positioning.

It is also of importance that crossover nulls in the off-axis frequency response be as narrow and unobtrusive as possible. This is usually assured by higher order network transitions with minimal overlap.

#### PERFORMANCE

A two-way monitor using the new horn and a 38 cm (15 in) nominal diameter woofer was designed which meets all the previously mentioned criteria (model 4430). Its performance in a variety of tests has been measured in comparison to previous designs of this and other companies.

The measurement most vital in revealing the performance of this type of monitor are the beamwidth/directivity curves. Use of a computer operated measurement system reduced the tediousness of these measurements and calculations (Keele [8]).

Figure 8 shows comparisons of the beamwidth and directivity of the 4430 (Fig. 8A) to two previous monitors. Our two-way 4331 model (Fig. 8b) had wide horizontal coverage but poor vertical coverage and constantly rising directivity (Q). The Q vs frequency of the coaxial design (Fig. 8c) appears smooth, yet the directivity index of the horn at higher frequencies is greater than the woofer by 4 dB, revealing a poor horn/woofer match. It is doubtful that a wide angle horn with good midrange control could be built into the available space of a typical coaxial design. Beamwidth vs frequency of this horn shows an interesting but undesirable trait, in that coverage angles vary in a complimentary fashion, which maintains consistent Q but does little to help off-axis response.

Figure 9 shows the normalized off-axis frequency response curves of the three systems. These curves represent those that would result from equalizing the axial response flat.

Power compression vs level is plotted in Figure 10. As the power levels were increased the chart recorder gain was decreased a like amount. The degree to which the curves coincide shows the system's freedom from the effects of compression. These curves were run using a narrow band tracking filter. The purpose of this tracking filter is to remove distortion components, which would otherwise influence the shape of these compression curves. As presented here, the compression curves reflect only the fundamental frequencies at each power level. Conventional distortion curves are also shown in Figure 11.

The group delay characteristics of the 4430, earlier 4331, and a popular constant group delay monitor are all plotted in Figure 12 vs the Blauert and Laws criteria for minimum audible time delay discrepancies [9]. Here, the constant group delay monitor excels, although all three easily fall well below the criteria.

For uses where even greater low frequency output capability with an attendant reduction in distortion is required, a double woofer system has been designed (model 4435). Directional characteristics have been left intact by bringing the second woofer in below 100 Hz only. The maximum output before thermal or excursion limiting has been raised by 4 dB and extended on the low end by half an octave. This is shown in Figure 13. Note that this is not a response curve but is instead a curve of maximum reverberant field SPL generated at the excursion limit or long term power limit of the two systems in typical monitoring conditions. In the very low-frequency range of 20 to 30 Hz a stereo pair of the dual woofer systems can generate some 115 to 120 dB SPL under these conditions.

Both low-frequency drivers in the double system are identical to the driver in the single system except for lightened cones, which yield a 3 dB increase in midband sensitivity.

#### RAMIFICATIONS OF THE NEW DESIGN APPROACH

##### 1. The room curve will be flatter; equalization will be more accurate

Studio monitors are generally equalized as a matter of course. Control rooms are rarely as smooth at low frequencies as may be desired, and mounting conditions for the monitors are not always ideal. Further, the user's concept of monitor equalization balance may not agree with that of the manufacturer.

Even though we believe that constant coverage monitors will require less equalization than previous designs, the need for equalization may still exist.

One of the more curious aspects of the recording art is the high frequency tailoring of playback monitors. Amplifiers are flat to one tenth of a dB; the ideal microphone is supposedly flat; tape recorders are aligned with great care in order to have as flat response as possible--yet control room monitors are traditionally rolled off, typically as much as 3 dB/octave above 4000 Hz (Schulein [10]). If this is not done, the response is often thought to be overly bright. Recent studies have shown that equalizing to such a rolled-off curve is merely a roundabout way of arriving at a flat direct sound field by allowing for the effects of increasing high-frequency directivity and decreasing reverberant field at high frequencies. In effect, we have been equalizing the reverberant field but listening to the direct field (Queen [11], Bridges [12]). When measured in the reverberant field the typical house curve exhibits a rolled off high end because previous monitors shared a similar power response rolloff. The degree of success with which a monitor could be properly equalized depended on its power response falling close to this "house curve." Deviations in the power response from this house curve would be equalized to yield complementary response errors in the direct field. For those cases where the power response did not properly follow the direct field response, equalizing would make the direct response worse, and hence degrade the perceived balance. The high degree of parallelism between the axial and power responses of the new monitor design means that less high end equalization will be required and, more importantly, that equalization will always be an improvement and never a degradation.

## 2. Stereo imaging will be improved

The frequency response of the new monitor design is quite uniform, even at angles sufficiently off axis both horizontally and vertically to be unlikely listener positions. However, this results in more uniform room reflections which contributes to a stable virtual source that does not change with frequency (Queen [13]). In addition, increased toe-in can be used with no degradation of the direct sound field. If enough toe-in is used for the axes of the systems to cross somewhat in front of the listener, then the level precedence effect can partially offset the time precedence effect (Haas [14]). This contributes to a more stable stereo image as the listener's position varies along the length of the control board.

## CONCLUSION

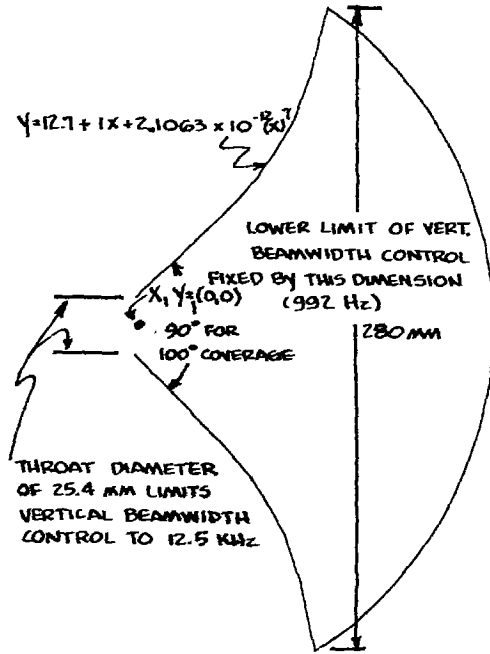
As all other parts of the recording chain are improved, the playback monitors must follow suit. Previous monitors had adequately flat axial frequency response and high acoustic output. By paying attention to power response and off-axis response, a monitor with fewer "colorations" and improved stereo effects can be realized. The use of a constant coverage horn allows the designer to create a two-way monitor that surpasses three- or even four-way monitors in several of these important aspects.

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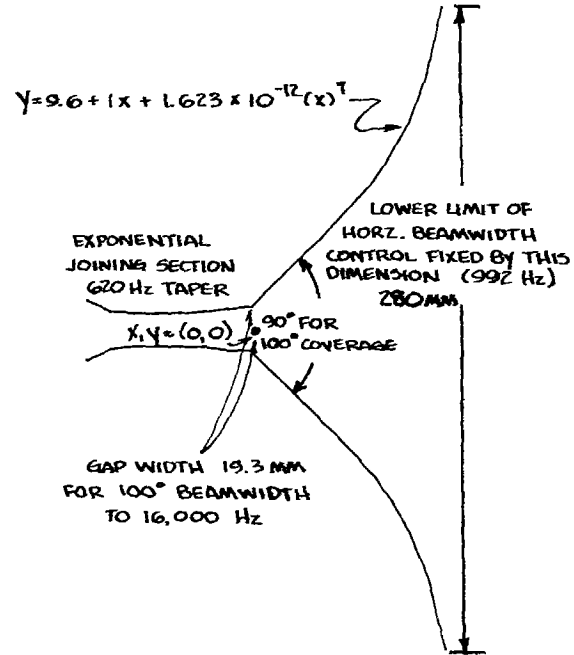
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### VERTICAL CROSS SECTION

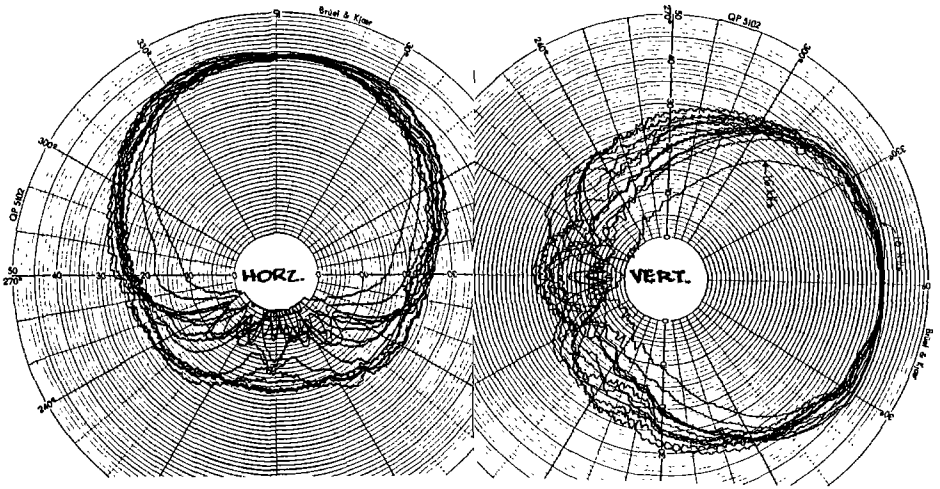


### HORIZONTAL CROSS SECTION



(ALL DIMENSIONS IN MM)

FIG. 1 KEELE CONSTANT COVERAGE BIRADIAL HORN - COMPUTER GENERATED PLOT



POLAR COMPOSITES 800-16,000 Hz

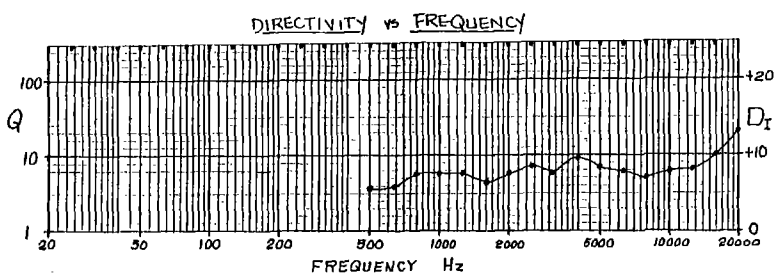
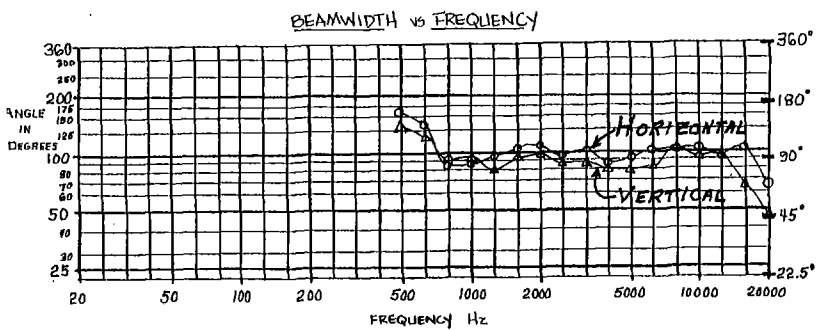
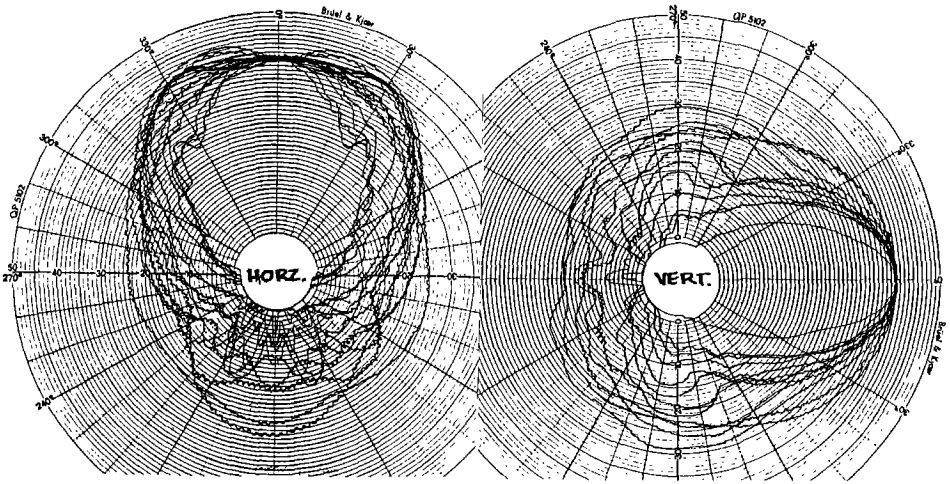


FIG. 2A BEAMWIDTH & DIRECTIVITY INDEX - CONSTANT COVERAGE HORN (234A)



POLAR COMPOSITES 800-16,000 Hz

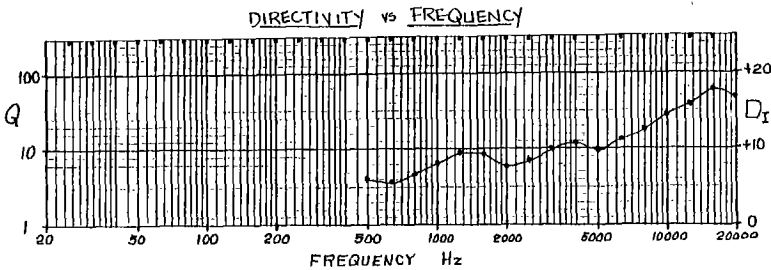
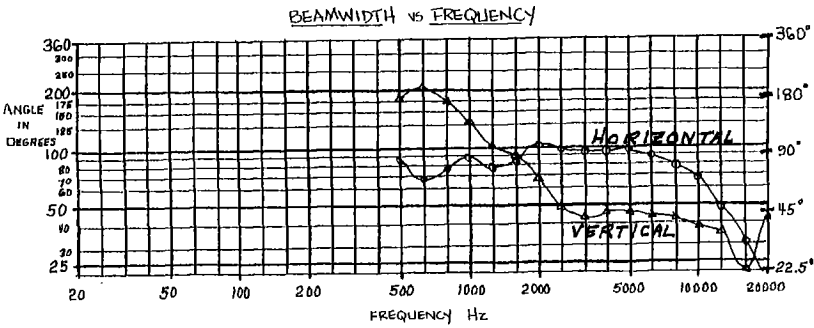
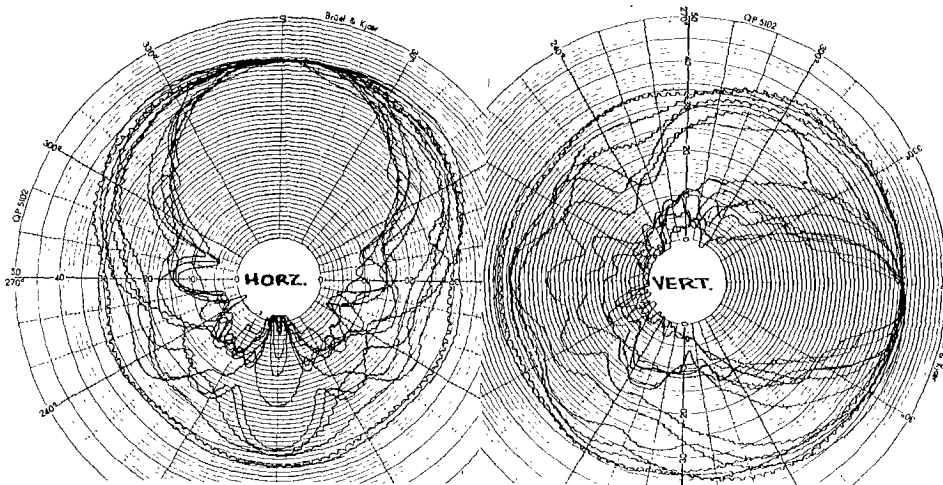
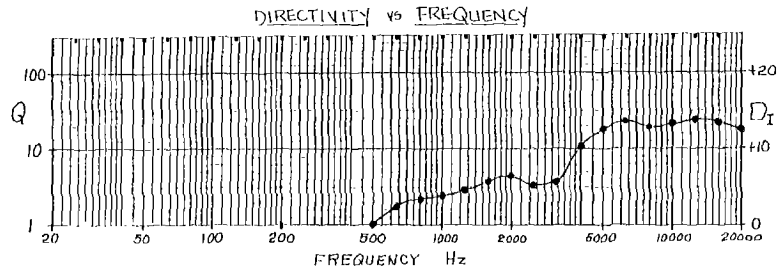
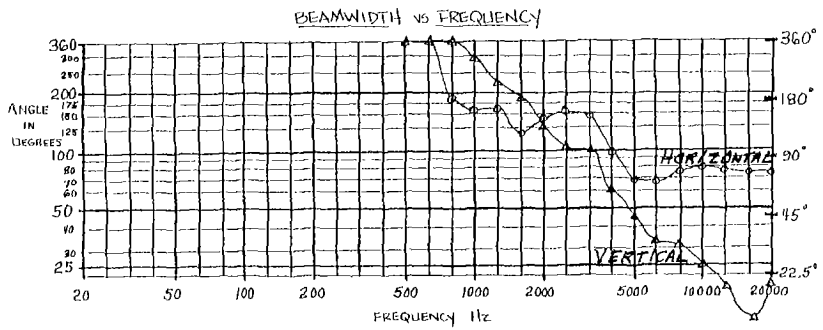


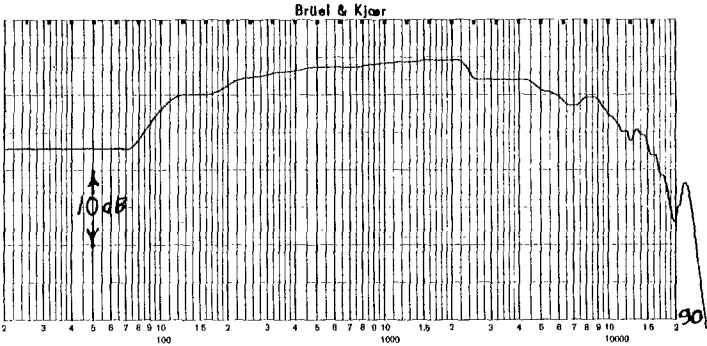
FIG. 2B BEAMWIDTH & DIRECTIVITY INDEX - RADIAL HORN (511B)



**POLAR COMPOSITES 800 - 16,000 Hz**

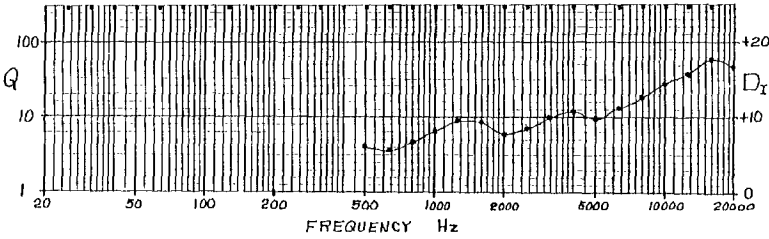


**FIG. 2C BEAMWIDTH & DIRECTIVITY INDEX - STRAIGHT EXPONENTIAL HORN AND DIVERGENT LENS (2307/2308)**

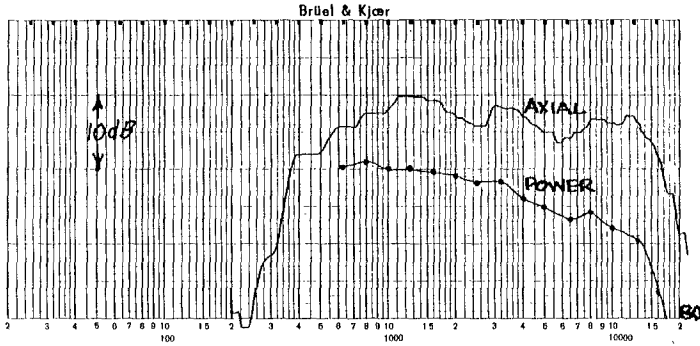


POWER RESPONSE OF COMPRESSION DRIVER (TERMINATED TUBE RESPONSE)

DIRECTIVITY vs FREQUENCY



DIRECTIVITY INDEX OF HORN

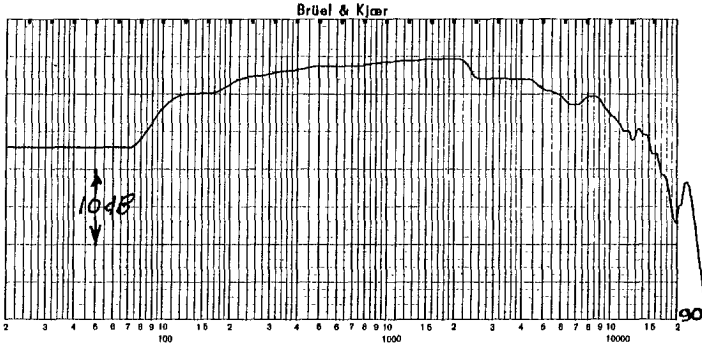


POWER CURVE =  
AXIAL CURVE -  $D_I$

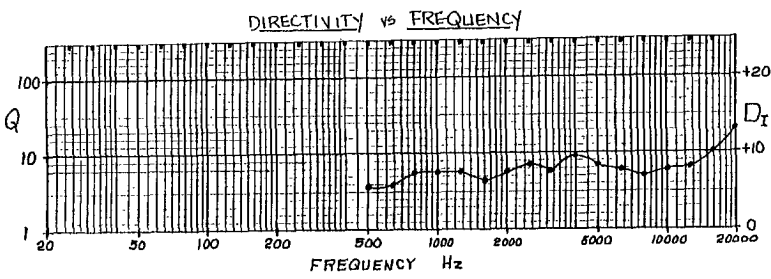
AXIAL CURVE IS  
FLAT BUT, POWER  
CURVE ROLLS OFF

RESPONSE, HORN AND DRIVER

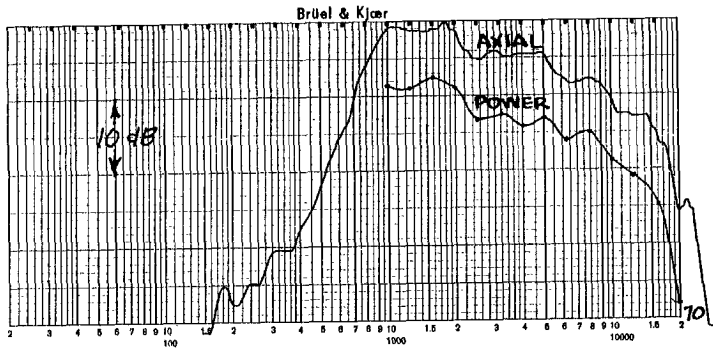
FIG. 3 ACOUSTICAL EQUALIZATION  
OF COMPRESSION DRIVER BY BEAMING HORN



POWER RESPONSE OF COMPRESSION DRIVER (TERMINATED TUBE RESPONSE)



DIRECTIVITY INDEX OF HORN



POWER CURVE =  
AXIAL CURVE -  $D_I$

POWER & AXIAL  
CURVE RUN  
PARALLEL

RESPONSE, HORN AND DRIVER

FIG. 4 AXIAL & POWER RESPONSE  
OF CONSTANT COVERAGE HORN

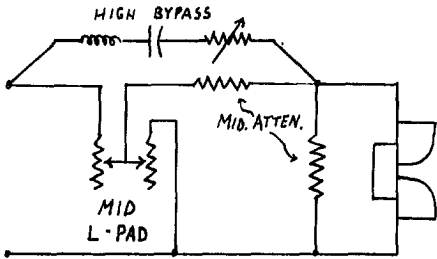
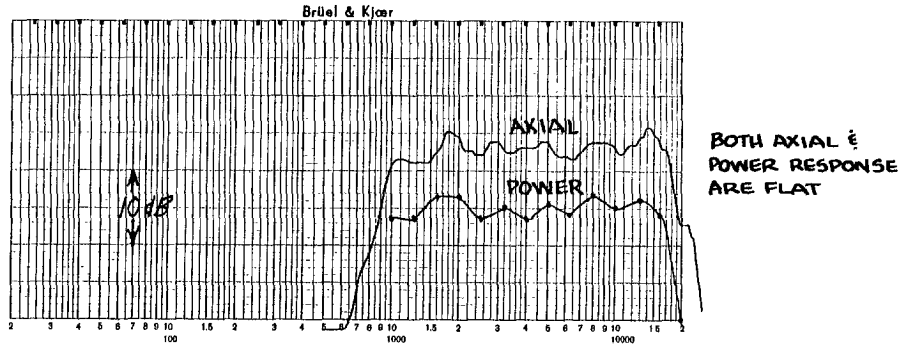
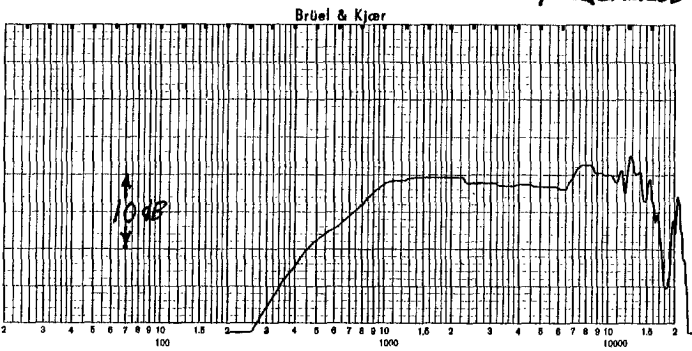


FIG 5 HIGH END E.Q. NETWORK



COMPRESSION DRIVER AND HORN ELECTRICALLY EQUALIZED



EQUALIZED COMPRESSION DRIVER ON TERMINATED TUBE-

FIG.6 EQUALIZED HORN RESPONSE,  
RESULTANT DRIVER POWER RESPONSE

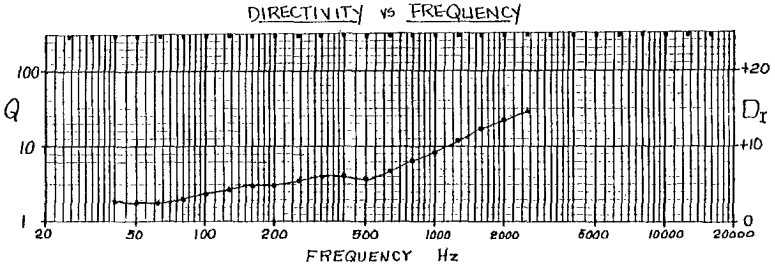
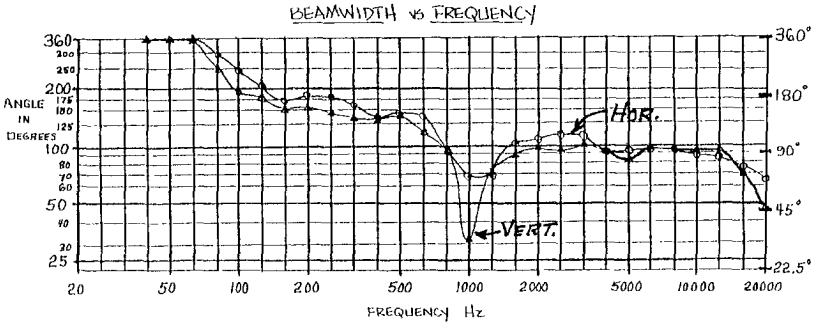
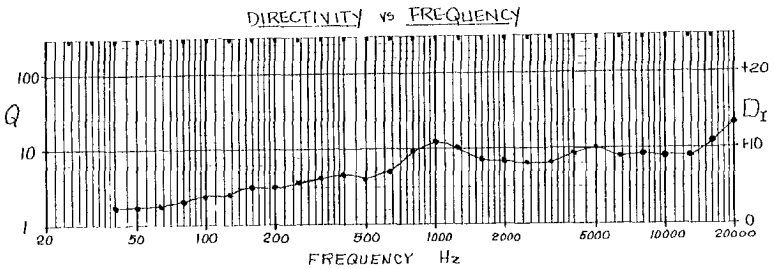


FIG. 7 DIRECTIVITY INDEX OF 38 CM WOOFER



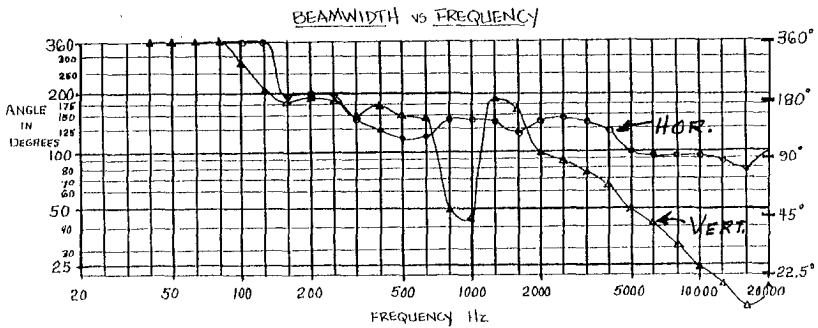
BEAMWIDTH



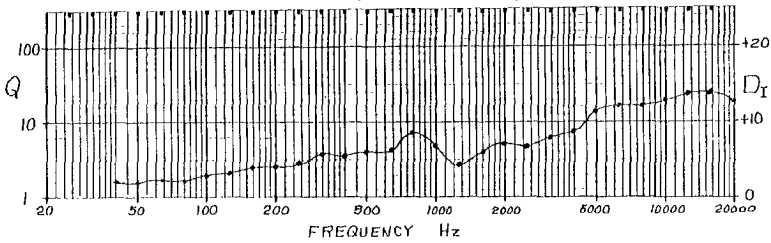
DIRECTIVITY INDEX

FIG. 8A BEAMWIDTH & DIRECTIVITY INDEX OF CONSTANT ENERGY RESPONSE MONITOR (JBL 4430)

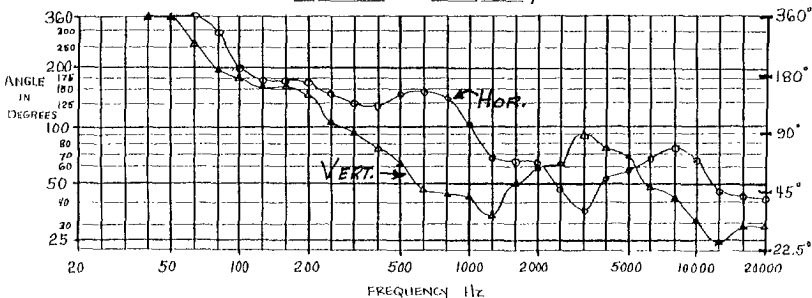




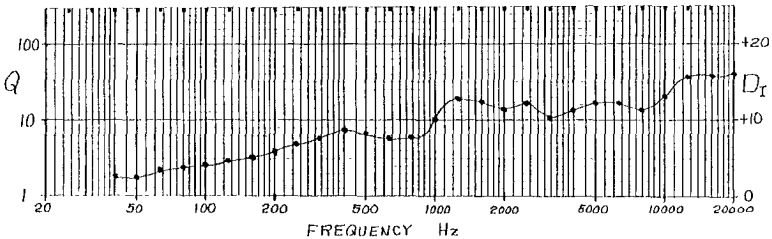
DIRECTIVITY VS FREQUENCY



**FIG. 8B BEAMWIDTH & D.I. (JBL 4331)**  
BEAMWIDTH VS FREQUENCY

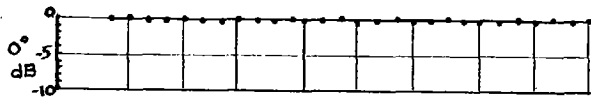


DIRECTIVITY VS FREQUENCY



**FIG. 8C BEAMWIDTH & D.I. - COAXIAL STYLE MONITOR**

IF THE 0° CURVE IS EQUALIZED TO FLAT:



THEN THE HORIZONTAL OFF AXIS CURVES ARE:

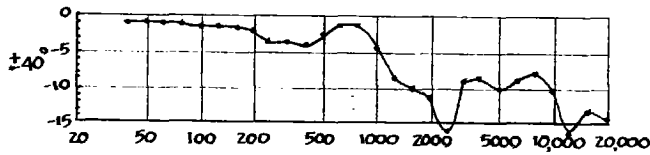
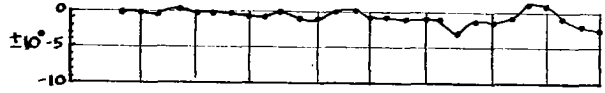
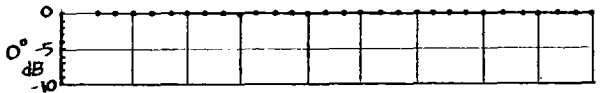


FIG. 9A POPULAR COAXIAL MONITOR

NOTE: EVEN COAXIAL MOUNTING OF HORN DOES NOT PREVENT OFF AXIS CROSSOVER DIPS; ALSO, IT EXHIBITS HIGH END BEAMING & RESPONSE DEVIATIONS DUE TO HORN.

IF THE 0° CURVE IS EQUALIZED TO FLAT:



THEN THE HORIZONTAL OFF AXIS CURVES ARE:

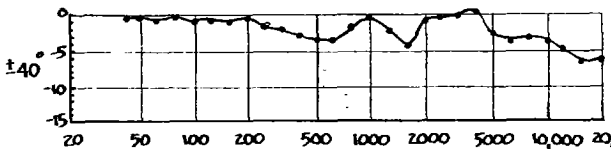
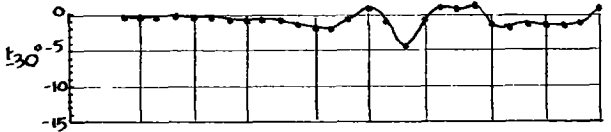
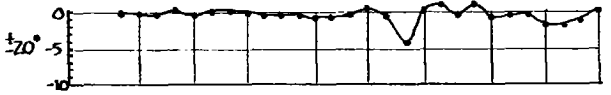
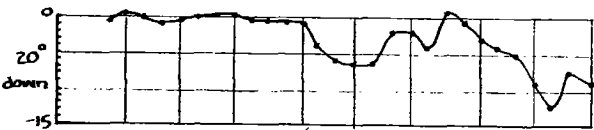
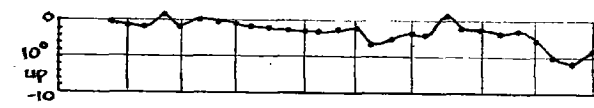


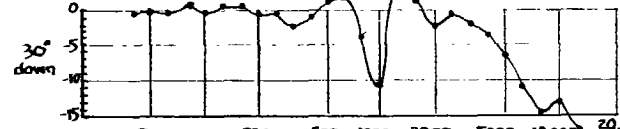
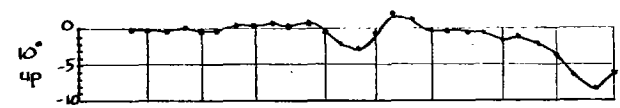
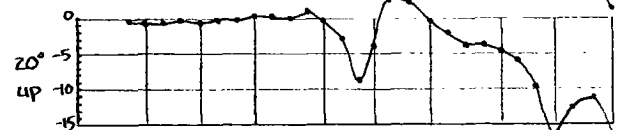
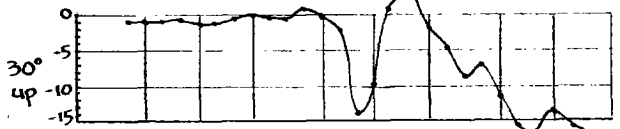
FIG. 9B 4331 MONITOR WITH ACOUSTIC LENS HORN

NOTE: GOOD HORIZONTAL CURVES BUT POOR VERTICAL CURVES WITH MUCH HIGH END BEAMING

AND THE VERTICAL OFF AXIS CURVES ARE:

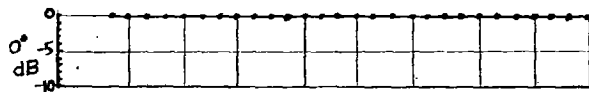


AND THE VERTICAL OFF AXIS CURVES ARE:

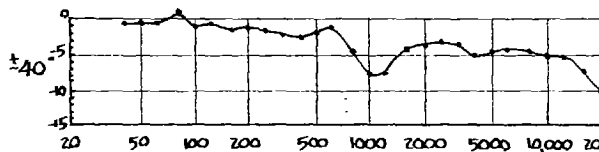
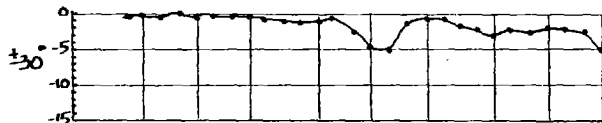
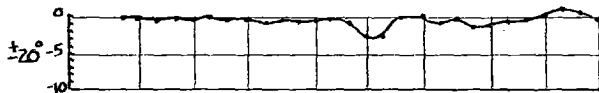
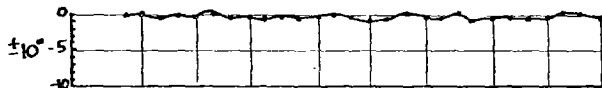


8713-10

IF THE 0° CURVE IS EQUALIZED TO FLAT:



THEN THE HORIZONTAL OFF AXIS CURVES ARE:



AND THE VERTICAL OFF AXIS CURVES ARE:

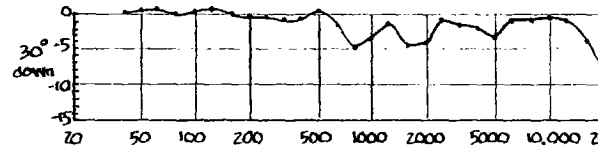
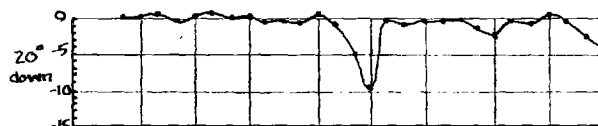
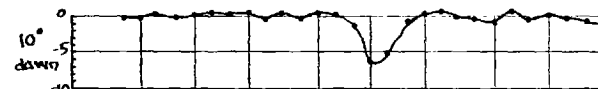
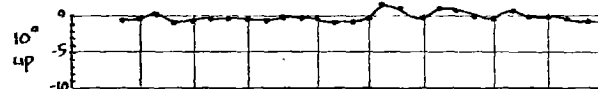
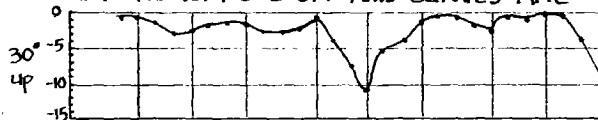


FIG. 9C 4430 CONSTANT COVERAGE MONITOR

NOTE: SMOOTH & EVEN CURVES WITH ABERATIONS CONFINED TO CROSSOVER REGION. VERTICAL RESPONSE OPTIMIZED FOR ON AXIS ± ABOVE (20° UP CURVE SURPASSES 10° DOWN CURVE).

Brüel & Kjær

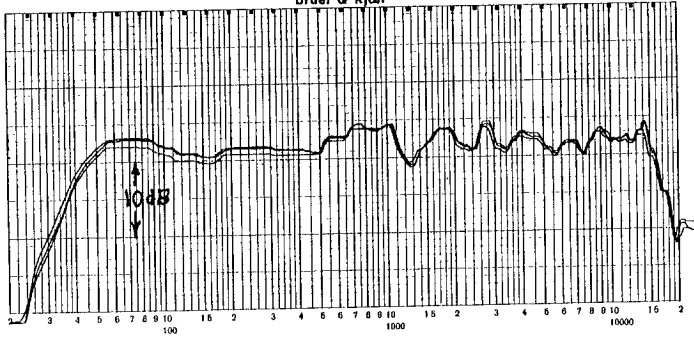
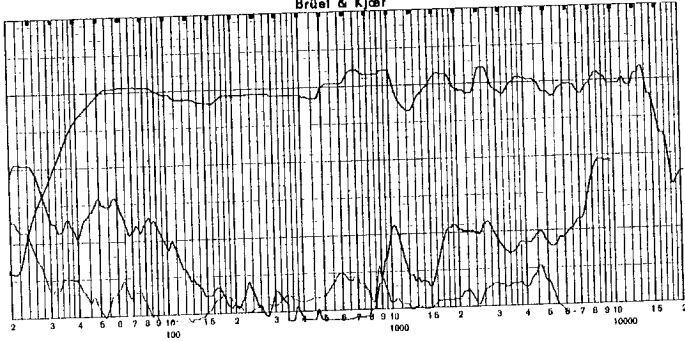


FIG. 10 POWER COMPRESSION 1, 10, 100 WATTS: 93, 103, 113 dB at 1 METER (4430 MONITOR)

Brüel & Kjær

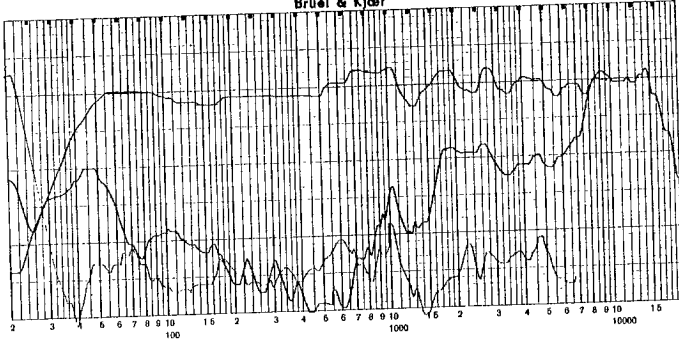


100 dB (5 WATTS)  
3.2%  
1.0%  
.32%

DISTORTION CURVES RAISED 20dB

DISTORTION COMPONENTS MEASURED TO 20,000 Hz ONLY

Brüel & Kjær



110 dB (50 WATTS)  
3.2%  
1.0%  
.32%

FIG. 11 DISTORTION VS FREQUENCY (2ND & 3RD HARMONIC) (4430 MONITOR)

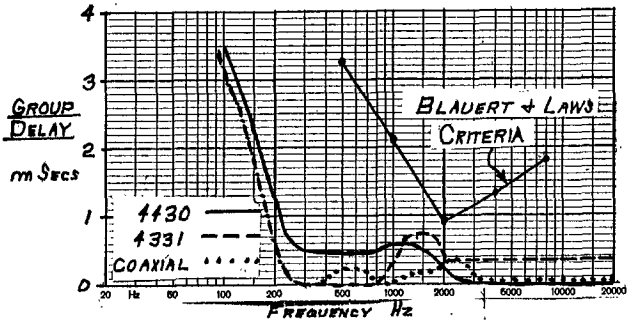


FIG. 12 GROUP DELAY VS. BLAUERT & LAWS CRITERIA

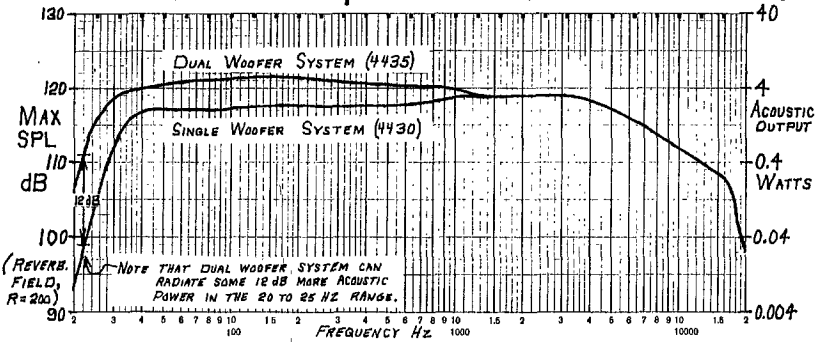


FIG. 13 MAXIMUM CONTINUOUS OUTPUT

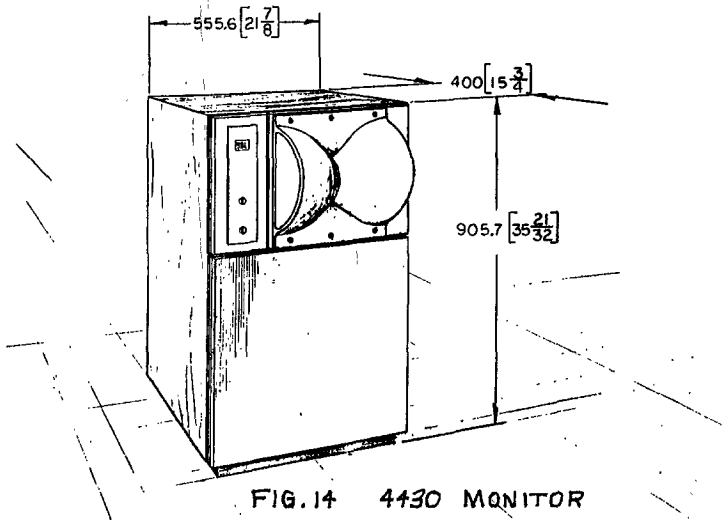


FIG. 14 4430 MONITOR