

KEF KEFTOPICS

A technical bulletin covering aspects of development, design and use of loudspeakers

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MAINTAINING QUALITY STANDARDS IN PRODUCTION

A great deal has been written about the development of new types of loudspeaker and the high standard of performance that can be achieved, but very little about all the checks and controls that are necessary to maintain this standard in commercial production. These processes must however be regarded as part of the overall system engineering strategy, since large production variations could render useless much of the effort put into the original design.

A modern high-quality loudspeaker, with its separate drive units covering different frequency bands, represents a system with many independent variables. The overall performance can be adversely affected by any mismatch between low-frequency, mid-range and high-frequency units, or by deviations of the component values in the crossover networks. To realise the full potentialities of such a system, a programme of rigorous production testing, covering all relevant parameters, is therefore necessary; each drive unit, for example, must be checked for conformity with the prototype on which the design was based, and matched, not only with other units in the same assembly, but with its opposite number in the other half of a stereo pair.

Some of these operations can now be carried out faster and more accurately by the application of digital techniques. A digital computer can derive the frequency response characteristics of a loudspeaker or drive unit from its impulse response, without the need for an anechoic environment*; it can also store the results from different tests and select or match units to within prescribed limits, all these processes being carried out within a time span short enough to satisfy the requirements of commercial production. Other tests likewise involve advanced instrumentation, and in general, meticulous attention to detail is necessary to ensure that the design concept is realised in every specimen of the final product. This issue of KEFTOPICS describes, as an example of modern practice, the various stages in the production test sequence of the KEF Model 105 loudspeaker.

*BERMAN, J. M. and FINCHAM, L. R.
(KEF Electronics Ltd.),
The Application of Digital Techniques to the
Measurement of Loudspeakers,
Journal of the Audio Engineering Society, Vol. 25, No. 6,
June 1977.

Drive Units

In the early days of large-scale loudspeaker manufacture, the mechanical properties of the diaphragm material were so variable that it was sometimes hard to find two drive units, even from the same batch, that sounded exactly alike. In more recent times, the situation has been ameliorated by the introduction of new materials, but close control is still necessary at all stages of production. In the Model 105, for example, the diaphragms of the low-frequency and mid-range units are vacuum-formed from Bextrene sheet, the mechanical properties of which have been measured, and coated with a visco-elastic damping material; they are then individually weighed, as shown in Figure 1, to check the thickness of the coating.

After assembly of the diaphragm with its voice coil — the resistance of which is kept to within $\pm 5\%$ of its nominal value — each drive unit is checked with a gliding tone for rattles or buzzes (an operation for which no satisfactory substitute for the human ear has yet been devised). The frequency of the fundamental resonance, i.e. that of the diaphragm moving as a whole on its suspension, is then measured on an “auto-resonator”, illustrated in Figure 2. This device takes advantage of the fact that the mechanical resonance of the unit produces a corresponding “motional impedance” component at the electrical input; the effect is the same as if a resonant circuit, consisting of an inductor and capacitor in parallel, had been introduced in series with the voice coil. The unit under test is connected to an amplifier in such a way that the motional impedance forms part of a positive feedback loop; the whole system then oscillates at the resonance frequency of the diaphragm, which is read off directly on a frequency meter. To avoid external influences, the units are tested unmounted so that the acoustic load on the diaphragm is small; however, the allowable variation in the resonance frequency as measured in this condition is based on its known effect on the response of the complete loudspeaker.

Acoustic Tests

The higher the standard of performance attained in the loudspeaker design, the greater the accuracy required in production testing to ensure that this standard is maintained. To do justice to the best modern reproducers, the acoustic response needs to be measured to an accuracy that until recently could be achieved — if at all — only by time-consuming operations in the open air or in a large and expensive anechoic chamber. This situation has however been radically altered by the new impulse testing technique referred to earlier.

When a very short electrical impulse is applied to the input of a loudspeaker, the resulting transient sound output consists of a complex series of decaying oscillations from which it is possible, in principle, to calculate the response to any other kind of input signal. In the past, however, attempts to derive such information have been beset by



Fig. 1. Weighing diaphragms to check the thickness of the visco-elastic damping layer; the balance is sensitive to changes of 10 milligrams.

practical difficulties — in particular by the poor signal-to-noise ratio at the “tail” of the transient — and only since the introduction of digital processing methods has it been possible to realise the full potentialities of this all-embracing technique.

The application of the new method is briefly as follows. The loudspeaker under test and the measuring microphone are set up in a room which excludes external noise but need not be completely anechoic. An electrical impulse is applied to the loudspeaker, and the waveform of the transient response picked up by the microphone is sampled at regular time intervals, at a rate fast enough to allow all the fine detail to be accurately analysed; the amplitude of each sample is converted to a digital number, and the resulting series of numbers is stored in a computer. In theory, the analysis should go on for ever, but in practice there comes a

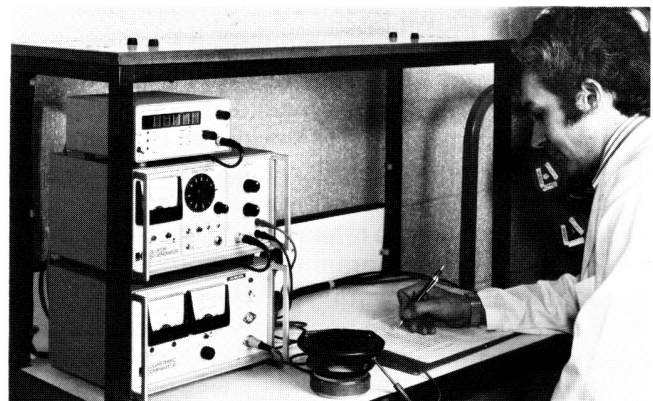


Fig. 2. Measuring the fundamental resonance of a drive unit; the “auto-resonator” makes the diaphragm vibrate at its natural frequency, which is displayed digitally.

point at which the sound has died away so far that the measurement can be terminated without appreciable loss of accuracy; the dimensions of the room should therefore be sufficient to ensure that this point has been reached, and all the necessary information has been captured, before any reflections from the walls, floor or ceiling arrive at the microphone. Rooms of somewhat smaller dimensions can be used, but as the reflections then arrive earlier, the latter part of the impulse response must be truncated to avoid interference, and a correction made in the analysis for the resulting modification of the low-frequency response.

The effect of any random noise in the measuring system is reduced by repeating the test a number of times at intervals just long enough to allow the room reflections resulting from one impulse to die out before the arrival of the next; in practice, this means that the minimum interval between impulses must be about equal to the reverberation time of the room, i.e. the time required for the sound to decay by 60 dB. Each new set of digital numbers — which represent the amplitudes of both the wanted signal and the unwanted noise — is added to those already stored, and the total divided by the number of tests to give the average. The improvement in signal-to-noise ratio comes about because each successive test produces an identical set of signal amplitudes, the running total of which thus increases in proportion to the

number of tests, whereas the corresponding noise amplitudes (which may be either positive or negative) add in random fashion, and their total therefore rises more slowly.

Finally, the stored and averaged impulse response is subjected in the computer to a mathematical process known as the Fast Fourier Transform (FFT); this converts the original function of time to a function of frequency, giving the amplitude/frequency response — and, if required, the phase/frequency response — that would have been obtained by a gliding-tone measurement made under ideal conditions. Experience has shown that the response of high-quality drive units such as those used in the Model 105 loudspeaker is of the “minimum-phase-shift” type; this means that the phase shift between electrical input and acoustic output is no greater than that inherent in the shape of the amplitude/frequency response characteristic. For production testing, only the frequency response, which gives the amplitude of the output without the phase, needs to be considered.

Acoustic tests of the kind described above are carried out at various stages in the production of the Model 105. Figure 3 shows one of these loudspeakers set up in a large impulse test chamber, which is primarily intended for research and design purposes, but is used also for the calibration of selected drive units that serve as reference standards. The dimensions of this chamber, 7.6 m x 7.6 m x 7.6 m, are large enough to allow the effective free-field response of a loudspeaker at a distance of 2 m to be accurately determined for frequencies down to 50 Hz. With no added sound absorbent material other than the carpet, the mean reverberation time, and hence the minimum interval between impulses is 4 seconds.

Production Testing

Figure 4 shows the set-up for batch testing of drive units. The units, previously stored at room temperature (18° to 20°C) for 12 hours, are mounted in turn on a flat baffle which is set flush in a bare wall of the test chamber, thus providing an unobstructed hemispherical radiation space. The floor, ceiling and remaining walls of the chamber are covered with a small amount of sound absorbent material which reduces the reverberation time over the frequency range of interest to 0.2 second, so that the impulse tests can be repeated at the rate of 5 per second. A total measurement time of only 30 seconds is sufficient to determine the true free-field frequency response to an accuracy of 0.3 dB. The dimensions of the chamber, 2.5 m x 3.15 m x 3.66 m, allow the free-field response of the units at 1 m to be measured down to 100 Hz; measurements at lower frequencies are not required at this stage, since the response of bass units below 100 Hz is determined by the fundamental resonance and the flux density, which have already been measured.

The drive units are tested in batches of 94; as a check, a calibrated reference unit of the same type

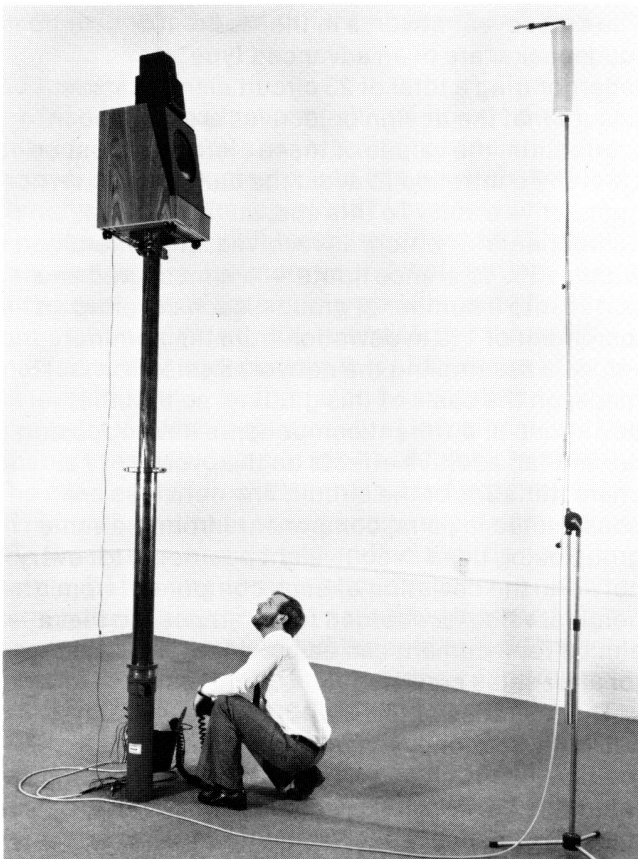


Fig. 3. Measuring the impulse response of a loudspeaker in a bare-walled test chamber; data required to compute the amplitude/frequency and phase/frequency response is picked up before first reflections arrive at the microphone.

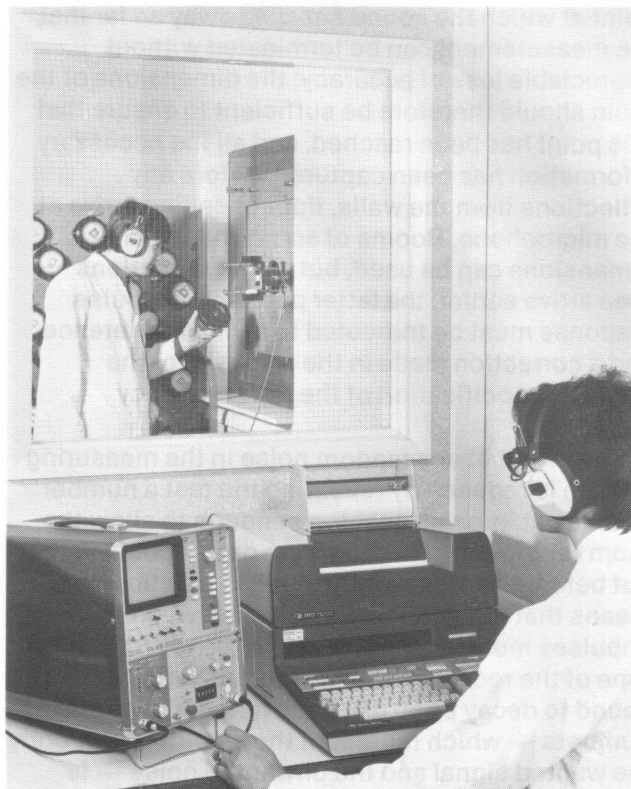


Fig. 4. Batch testing of drive units by impulse measurement.

is measured at the beginning and end of each set of tests. The impulse response of each unit is stored on magnetic disc; from this information, the computer calculates the corresponding frequency response and then carries out a matching programme for the batch, listing the sensitivity of each unit with respect to the reference unit together with the maximum deviation of the unit response — measured at eight frequencies per octave — with respect to the reference response. (If required, a printout can be produced showing, for each of these frequencies, the statistical distribution of the results.) The computer programme also matches the units into pairs, giving the mean sensitivity for the pair and the residual match error, if any; Figure 5 shows a typical computer printout of a matching routine together with the frequency response characteristics of two units paired by the computer with a match error of 0.5 dB. Finally, sets of six matched units are selected to make up pairs of matched loudspeaker systems; Figure 6 shows sets of selected units en route to the assembly area, illustrated in Figure 7. To facilitate after-sales service, test results for all drive units are recorded, so that in case of accidental damage, a pair of replacement units can be supplied which are matched, not only to each other, but to the other units in the system; the relevant information for each pair of loudspeakers is filed in the form of a "customer printout", an example of which is reproduced in Figure 8.

Crossover Networks

The crossover networks in the Model 105 loudspeaker are of an advanced type*, incorporating a total of 23 circuit elements; to ensure that the design objectives are realised in production, the values of these elements must be carefully controlled to avoid the danger of cumulative errors. To this end, all the network components — which, as received, are generally within $\pm 5\%$ tolerance limits — are measured and sorted into a number of groups, each covering an increment of 1% in deviation from the nominal value; in assembling the networks selection is made, on the basis of this grading, so that the deviations of different components have opposing rather than additive effects on the overall characteristics of the circuit. The networks are constructed in pairs, components from the same group being used in equivalent positions; for every network, the deviation of each component from its nominal value is recorded for reference, so that a similar replacement can be provided, if necessary, for after-sales servicing.

The response of the low-frequency, mid-range and high-frequency network assigned to each loudspeaker is checked against a corresponding reference network by a sensitive comparator, the maximum disparity allowed being $\pm \frac{1}{2}$ dB over the pass-band in each case, and ± 1 dB in the cut-off region; the two corresponding networks of a

*KEFTOPICS Vol. 3, No. 1.

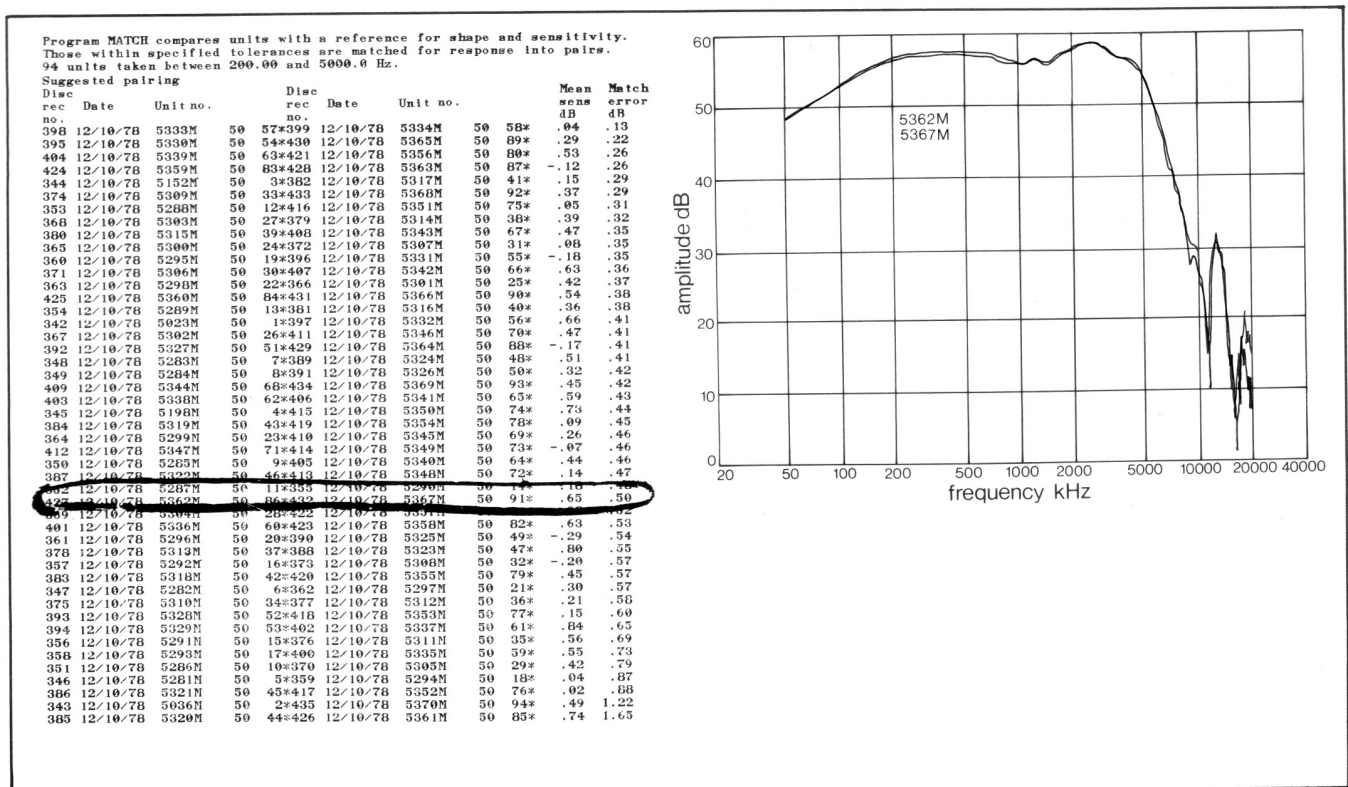


Fig. 5. Computer printout from the matching routine for a batch of 94 drive units, together with the frequency response curves of a typical pair of matched units.

loudspeaker pair are similarly checked against each other, and must not differ by more than 0.2 dB.

Head Assembly

In the Model 105 loudspeaker, the mid-range and high-frequency drive units, with their associated crossover networks, are housed in a separate head assembly mounted on top of the bass enclosure. The head assembly, which can be turned and tilted as required to give optimum sound distribution over the designated listening area, is provided with a light-emitting diode (LED), energised by the audio signal fed to the loudspeaker; this LED serves as a peak level indicator, but can also be switched temporarily to glow continuously at low signal level to enable the positioning of the head to be checked.

The bass enclosures are constructed in pairs with matching walnut veneer, and to avoid damage to the finish, they should be handled as little as possible. Before the complete loudspeaker is put together, therefore, the head assembly is tested separately while mounted temporarily on a dummy enclosure to simulate the final working condition. The gliding tone test for rattle and buzz, already carried out on the individual drive units, is repeated for each head assembly, and the operation of the LED is checked. The acoustic frequency response is then measured; each head assembly, mounted on its dummy bass enclosure, is carried on a roller conveyor into a system test chamber, illustrated in Figure 9, where a trolley driven by a linear induction motor brings it to the measuring position.

The dimensions of the system test chamber, 5.8 m x 4.8 m x 3.66 m, are large enough to allow measurements to be made on loudspeaker assemblies at the typical listening distance of 2 m. As before, sufficient sound absorbent material — distributed in this case over all six surfaces — is provided to reduce the reverberation time to 0.2 second, so that tests can be repeated at the rate of 5 impulses per second. The computer stores the characteristics of the two head assemblies designated as a stereo pair, and calculates the difference between them; both these characteristics and their difference — which must be less than 1 dB — are then drawn out on an X-Y plotter for inspection.

Overall Test

In the last stage of construction and objective testing of the Model 105, each head assembly is mounted on a bass enclosure fitted with the low-frequency drive unit already assigned to it, and the complete loudspeaker is given a further glide-tone check for rattle and buzz. The impedance/frequency curves for the two loudspeakers of a stereo pair are then traced, using an X-Y plotter coupled to an oscillator; the impedances must not fall below the specified minimum, or differ from one another by more than 0.5 ohm. Finally, each loudspeaker is returned to the system test chamber of Figure 9 where the impulse response is measured at 2 m on the design axis. The resulting data is stored on a magnetic disc for future reference, while the

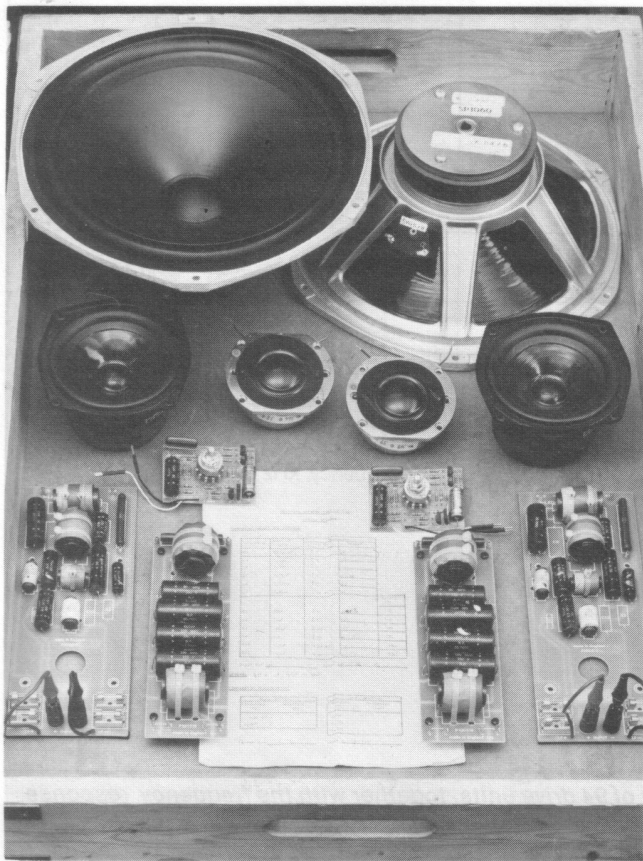


Fig. 6. Drive units and crossover networks in matched sets ready for assembly.

corresponding frequency response, from 50 Hz to 20 kHz, is plotted out; the difference between the two loudspeakers of a pair is also plotted, using a decibel scale magnified ten times so that even small disparities can be observed. The impulse response is truncated, as described earlier, to eliminate room reflections, and to obtain the true free-field characteristics below 150 Hz, a constant correction is made in the computer.

The tolerances applied to the overall frequency response are in practice more stringent than might be expected from the ± 2 dB limits given in the loudspeaker specification. Those limits, if

KEF MODEL 105 REFERENCE SYSTEMS				
ASSEMBLY RECORD FOR SYSTEMS NO. 3187A & 3187B				
CUSTOMER COPY				
ITEM	MODEL	TYPE	SERIAL NOS.	
LOUDSPEAKER SYSTEM	105	SP1059	3187A	3187B
HIGH FREQUENCY UNIT	T52	SP1049	5162	7392
MID FREQUENCY UNIT	B110	SP1057	5106	5207
LOW FREQUENCY UNIT	B300	SP1060	4949	4986
CROSSOVER NETWORK	HF/HF	SP1100	5458	5459
CROSSOVER NETWORK	LF	SP1101	4888	4889
PEAK LEVEL INDICATOR		SP1062	4971	5005
A ASSEMBLED BY: 2, 3		B ASSEMBLED BY: 2, 3		
A INSPECTED BY: 7, 7		B INSPECTED BY: 7, 7		
15/10/78				
KEF ELECTRONICS TOVIL MAIDSTONE KENT ENGLAND				

Fig. 8. Customer printout showing serial numbers of drive units, crossover networks and peak level indicators fitted to one loudspeaker pair.

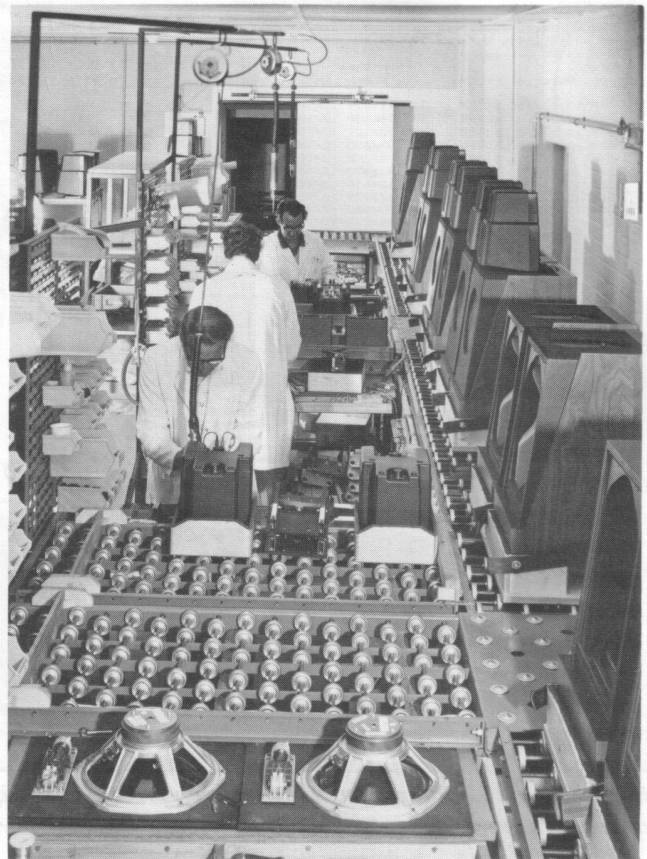


Fig. 7. General view of assembly area for Model 105 loudspeakers.

legally interpreted, would allow, for example, a continuously sloping characteristic with the top end 4 dB above the bass, or one that lies mainly on the lower boundary but has a number of 4 dB peaks. KEF, however, work on the principle that the frequency response of production models should be kept as close as possible to that of the prototype — which has already been shown to be acceptable — and that the most realistic criterion is therefore the deviation from the design centre curve. By rigorous testing and matching of components in the earlier stages of production, the spread in the axial frequency response of the Model 105 has been kept within the limits shown in Figure 10. The dashed curves are spaced by two standard deviations* above and below the design centre response; just over 95% of the characteristics measured on the production line lie between these curves, and thus fall well inside the ± 2 dB specification. In addition, as a result of the selection of components, the frequency characteristics of any two loudspeakers designated as a stereo pair are closely matched, as illustrated by the typical example shown in Figure 10a.

*The standard deviation is a measure of variability; it is the root-mean-square value of the amounts by which the results for the individual specimens tested differ from the average.



Fig. 9. System test chamber for head assemblies and complete loudspeakers.

Subjective Tests

After the comprehensive series of objective tests already applied to each loudspeaker, subjective assessment of performance might appear superfluous; experience shows, however, that a last-minute check of this kind is advisable, if only to detect faults arising, for example, from accidental damage or disconnection of units.

Each pair of loudspeakers is set up in the room illustrated in Figure 11, which is designed to provide ideal listening conditions in a room of typical domestic size, and has a mean reverberation time of 0.25 second. Initially, the two systems are placed side by side and fed with identical signals. The LEDs are checked for equal brightness, first in

their low-level operating mode, using continuous pink noise, and then in their normal function as peak level indicators, using high-level pink noise pulsed to simulate programme.

The listening tests begin with pink noise, presented alternately on the left and right channels as a sensitive check on the tonal balance of the loudspeaker pair. This is followed by recorded speech, again alternating between the two loudspeakers; the voice is that of a male member of staff well known to the test engineer, and therefore constitutes a critical test for naturalness. Subjective assessments of this kind can give misleading results if the volume of the reproduced

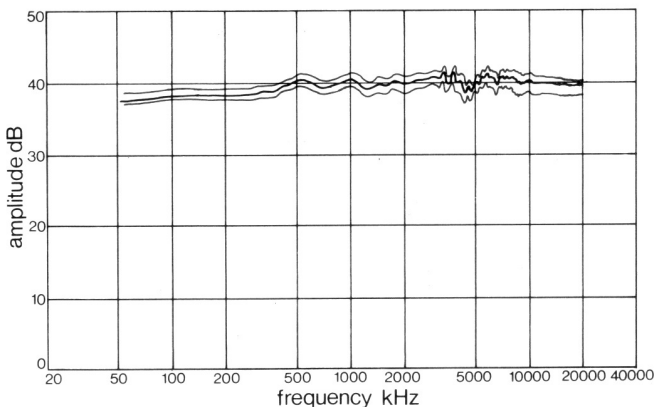


Fig. 10. Axial frequency characteristic of Model 105 loudspeaker.

— design-centre response
 — production spread ($\pm 2 \times$ standard deviation)

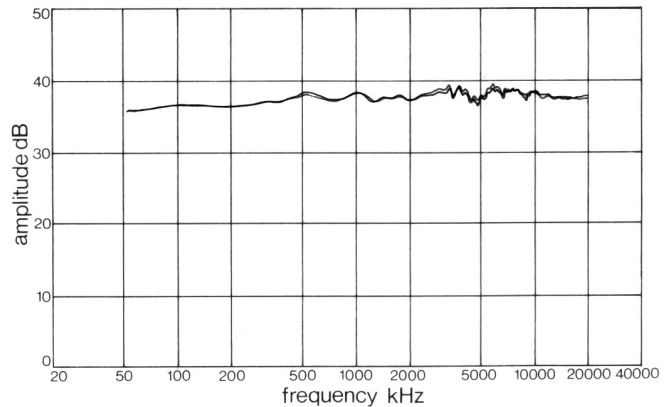


Fig. 10a. Frequency response curves of a typical matched pair of Model 105 loudspeakers.

sound is too high or too low; all signal levels are therefore accurately standardised.

Finally, the two loudspeakers are set up at the normal spacing for stereo reproduction, as shown in the figure, and recordings covering a broad cross-section of music, including strings, brass, piano and other percussion instruments are replayed.

Assuming that nothing untoward is observed — and it is indeed rare for any fault to be found at this late stage — the test schedule concludes with a minute visual examination, under a strong light, for finger marks or other blemishes that may have resulted from handling and must be removed before the packaging stage.

Conclusion

Modern practice in the production of professional-grade loudspeakers such as the Model 105 is far in advance of the methods adopted a few years ago. In those days, drive units were selected on the basis of relatively coarse measurements, the grosser variations in overall system response being mitigated by the use of “adjust-on-test” elements in the associated electrical networks. By contrast, the production schedule outlined above is based on measurements carried out with an accuracy and thoroughness formerly possible only under laboratory conditions, combined with a comprehensive process of grading and matching; as a result, the performance of the end product can be held, without the need for adjustment, within limits close enough to satisfy the most stringent requirements — for example, those imposed on loudspeakers designed for monitoring purposes.

Although many of the operations described depend on computer technology, and in principle involve some advanced theoretical concepts, their practical application in a production routine is simple and straightforward. The cost of the necessary equipment has amply justified itself, not only by the improved performance of the product, but by the low fault rate that comes from stringent testing, plus the effective after-sales service made possible by the keeping of detailed records; as a result of the experience with the production of Model 105, it has been decided to apply the new methods to future KEF loudspeakers.

Because of the high degree of consistency obtainable by modern technology, it can now be claimed with confidence that the standard of performance achieved by the designer is not confined to the prototypes used in demonstrations and press reviews, but is realised in full for each and every purchaser.



Fig. 11. Listening room for subjective checking. The sound absorption units on the walls and ceiling are designed to provide ideal listening conditions.

